



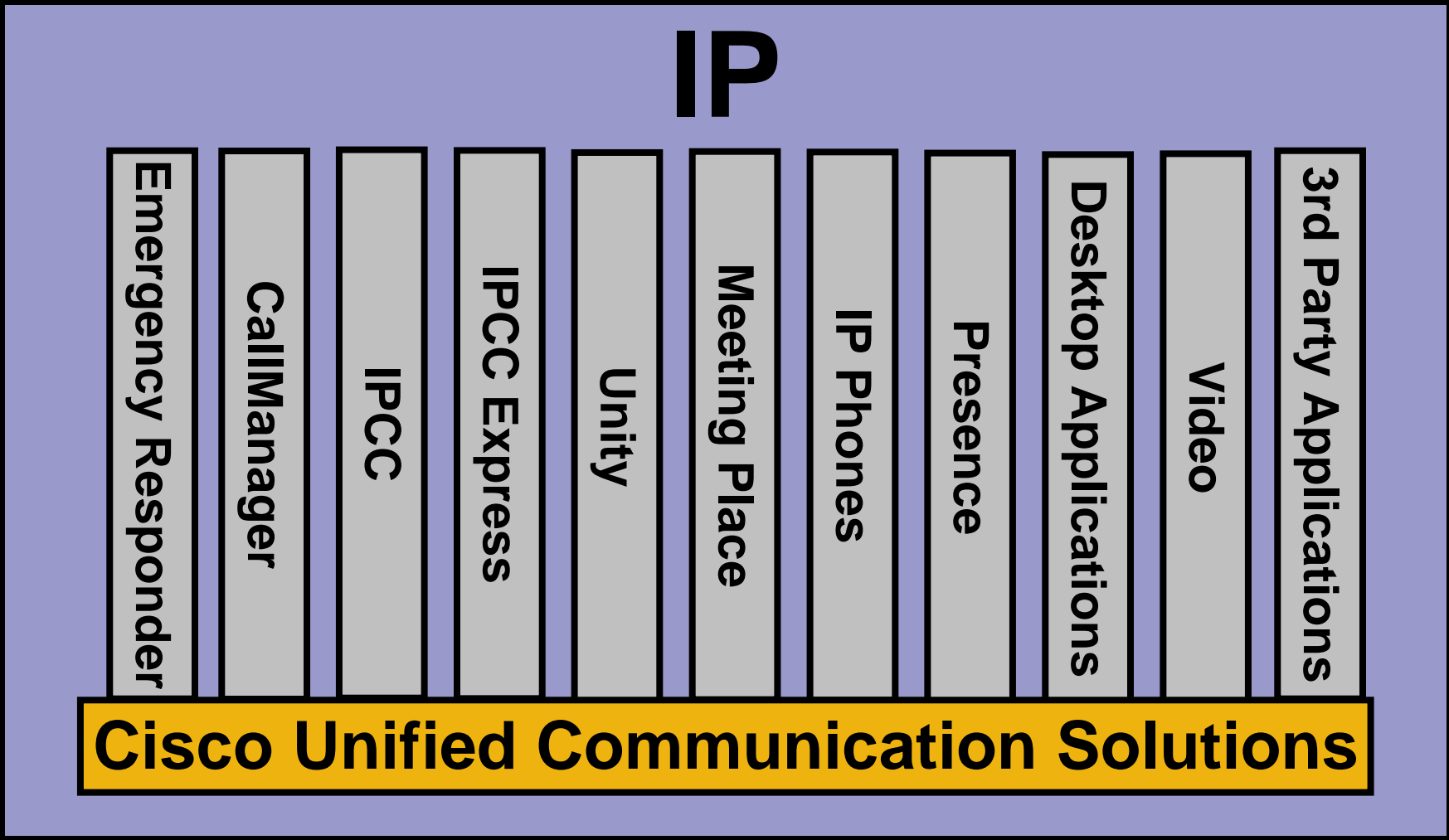
## **Understanding Cisco Enterprise IP Communication**

# **Introducing Cisco Enterprise IP Communication Components**

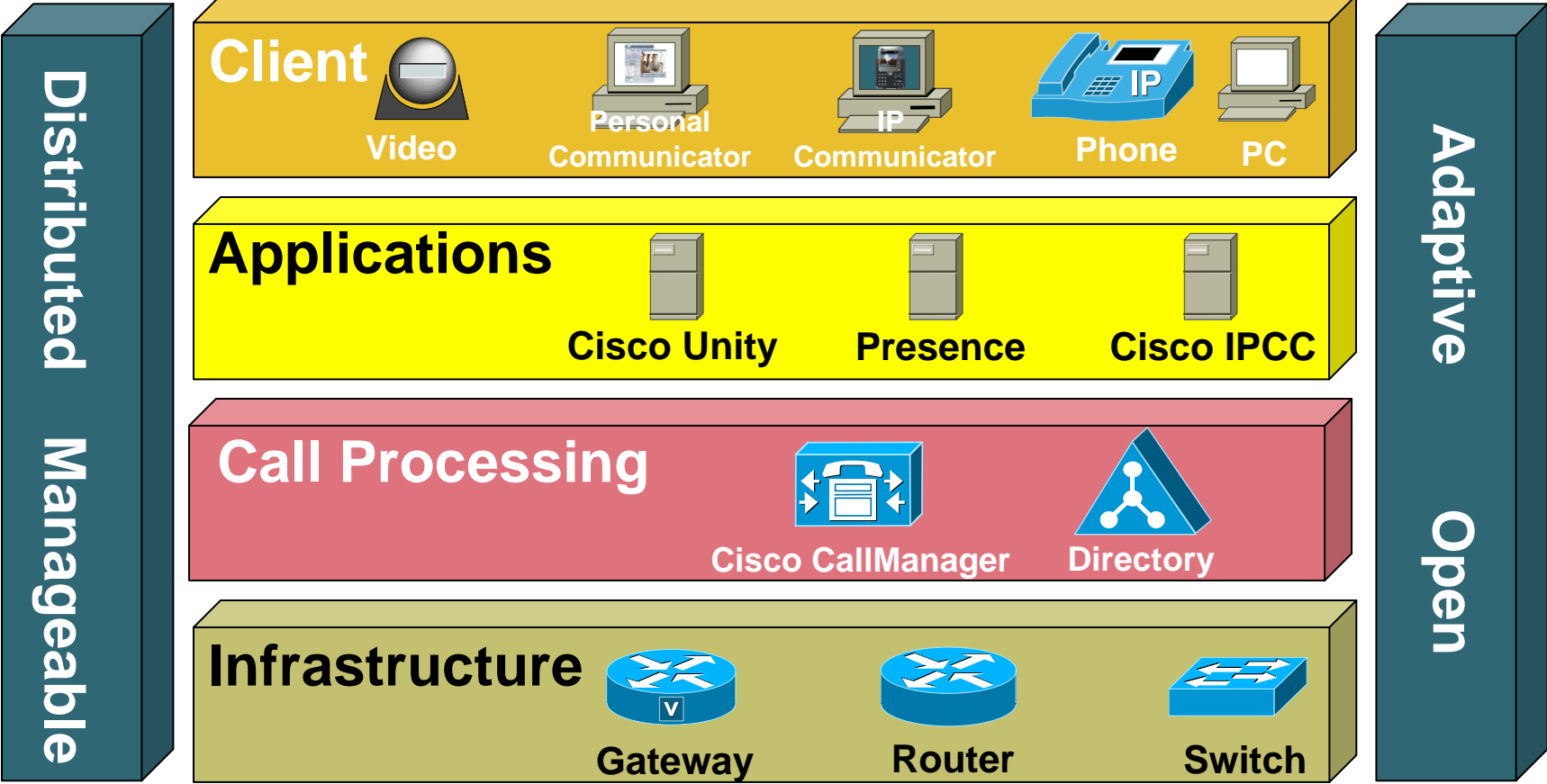
# Objectives

- **Cisco Unified Communications Overview**
- **Cisco IP Communication Components**
- **Columbus vs. Golden Bridge**
- **Cisco Unity**
- **Cisco IP Contact Center**
- **Cisco Unified Presence Server**

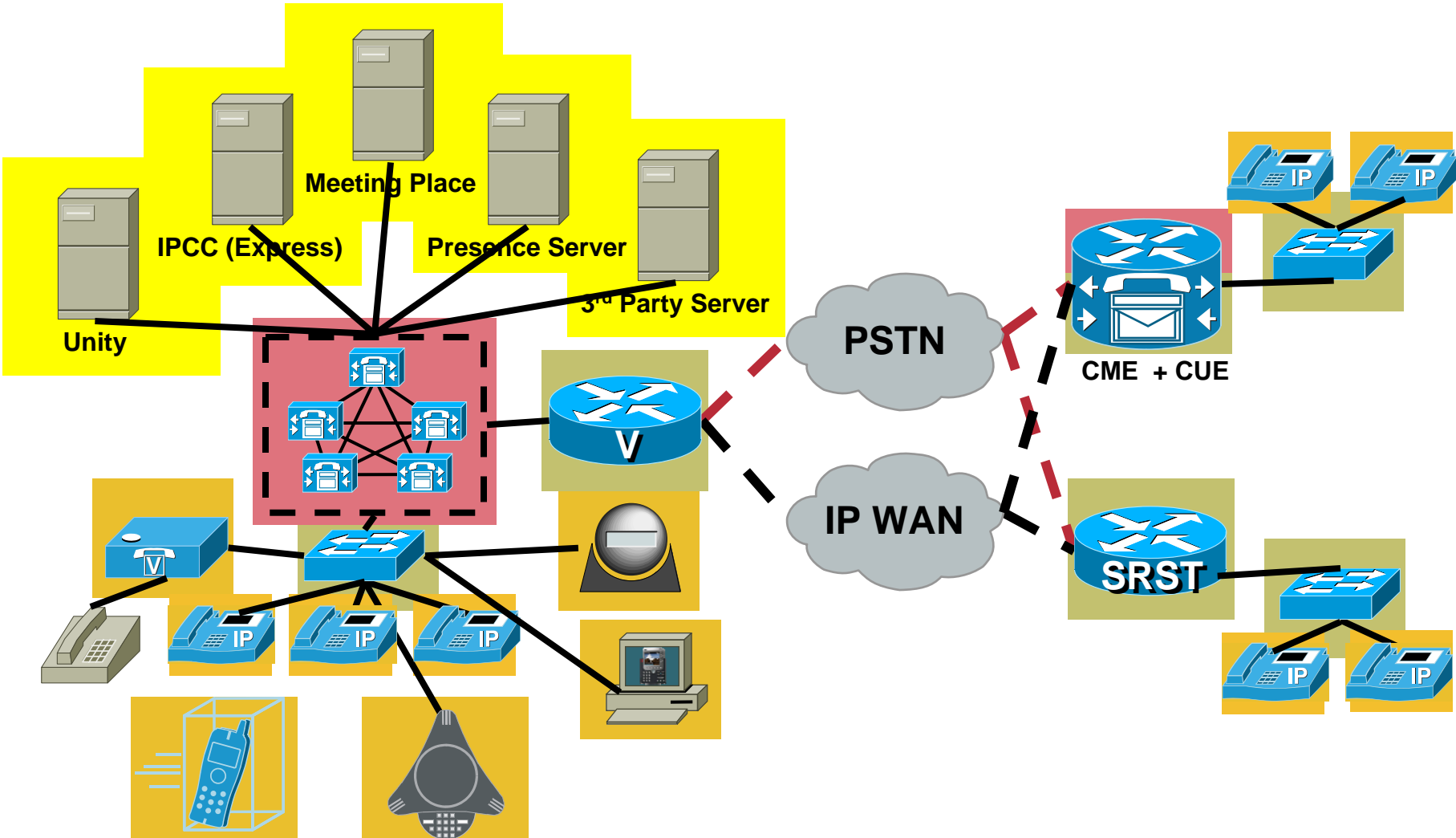
# Cisco Unified Communications Overview



# Cisco Unified Communication Layers



# Cisco IP Communication Components



# The Columbus Mission

**Represents a new way of delivering Cisco Unified Communications functionality to enterprise customers.**

- **Columbus integrates telephony, conferencing (voice, video and web), messaging, and contact center technology.**
- **Centered on the CCM 5.0 release.**
- **The first coordinated system release of an integrated set of products tested, documented and supported as a system.**

# The Columbus Mission (cont.)

- **Delivers a collection of products with one release date, testing methodology, and documentation.**
- **The first in a series of time-based releases integrating CCM with the current versions of all IPC products.**
  - **Unity, IPCC, CRS, Customer Voice Portal, MeetingPlace, NextGen IP Phones, phone and desktop client applications, etc.**
- **Columbus is the successor to Golden Bridge.**

# A Review of Golden Bridge

## System Testing for Enterprise IPC

- **Components = Infrastructure, Gateways, Call Control, Communications Applications, Management, Security**
- **Validates system compatibility, functionality, load etc.**
- **Integral part of Enterprise Voice solution management**

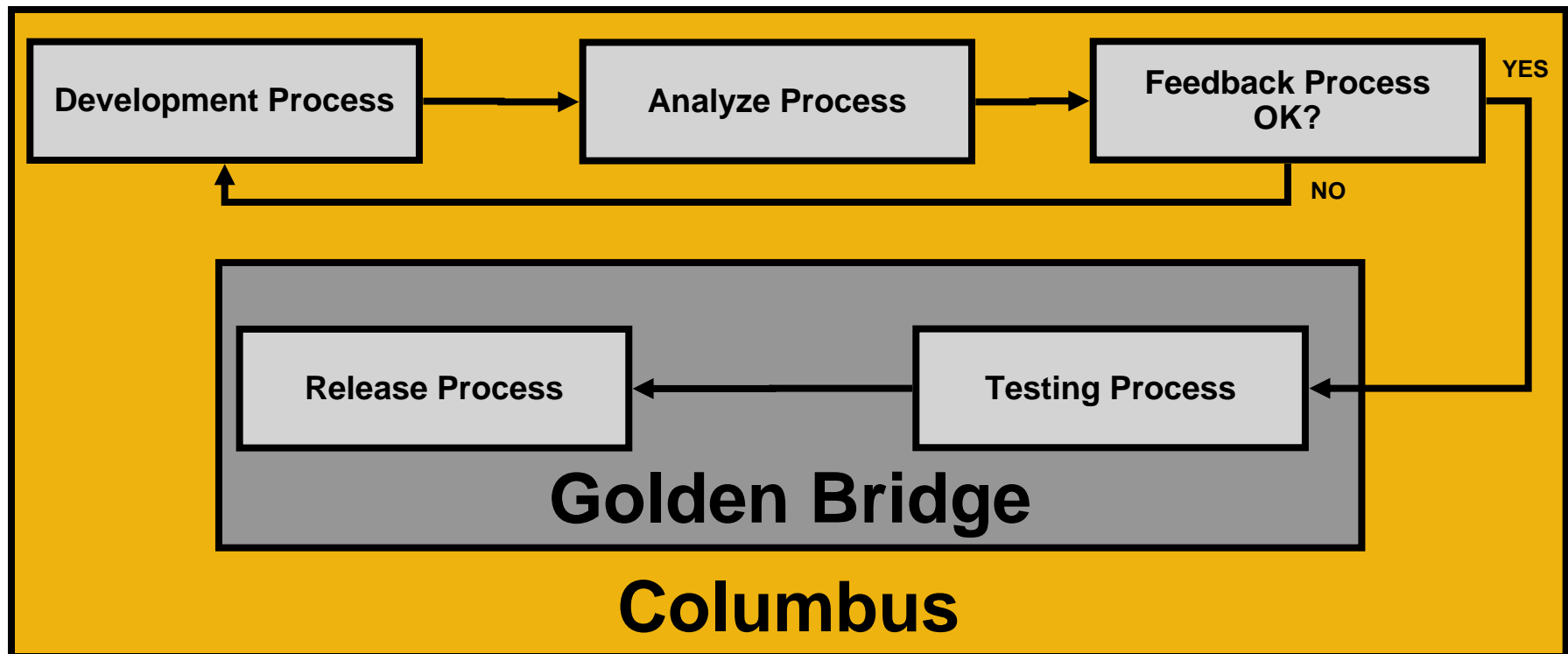


**The goal of Golden Bridge is to ensure success of Unified Communications systems throughout their lifecycle**



# Columbus Testing, Design, and Implementation Process

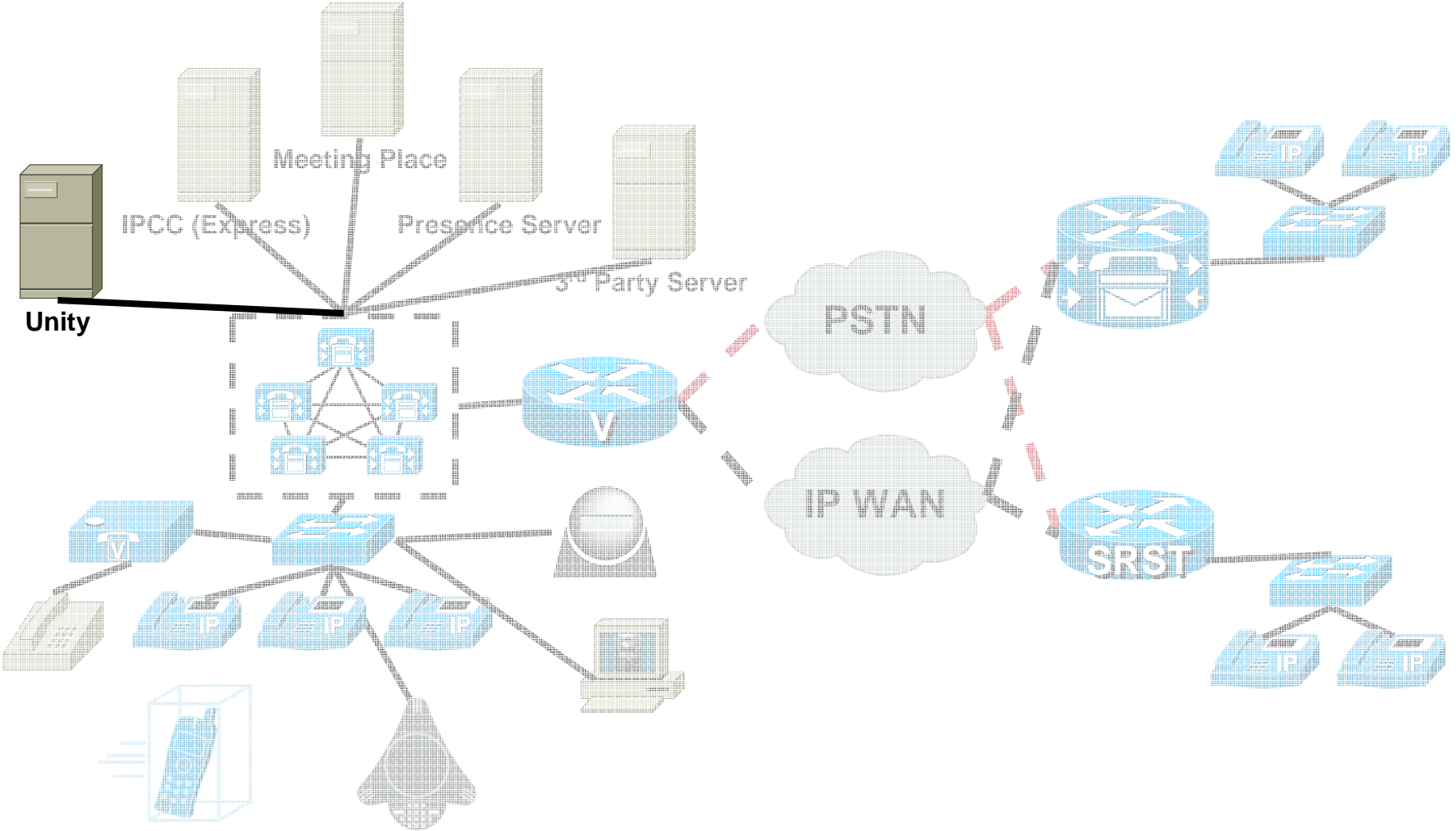
**Columbus is a wider approach that includes the Golden Bridge process.**



# Columbus versus Golden Bridge

Function	Golden Bridge	Columbus
System Development Requirements		<b>Partial</b>
System Functional Spec		<b>Partial</b>
System Test Requirements	<b>Yes</b>	<b>Yes</b>
Integration	<b>Yes</b>	<b>Yes</b>
System Testing	<b>Yes</b>	<b>Yes</b>
System Documentation	<b>Yes</b>	<b>Yes</b>
System Performance Characterization	<b>Yes</b>	<b>Yes</b>
System EFT	<b>Partial</b>	<b>Yes</b>
System Training		<b>Yes</b>
System Marketing Launch		<b>Yes</b>
System Orderability		<b>Yes</b>
System-wide Licensing and Pricing		<b>Yes</b>

# Cisco Unity



# Cisco Unity

**One-stop message access *anywhere, any way, anytime.***

**Broad range of options for advanced productivity with integrating voicemail, unified messaging & personalized productivity.**

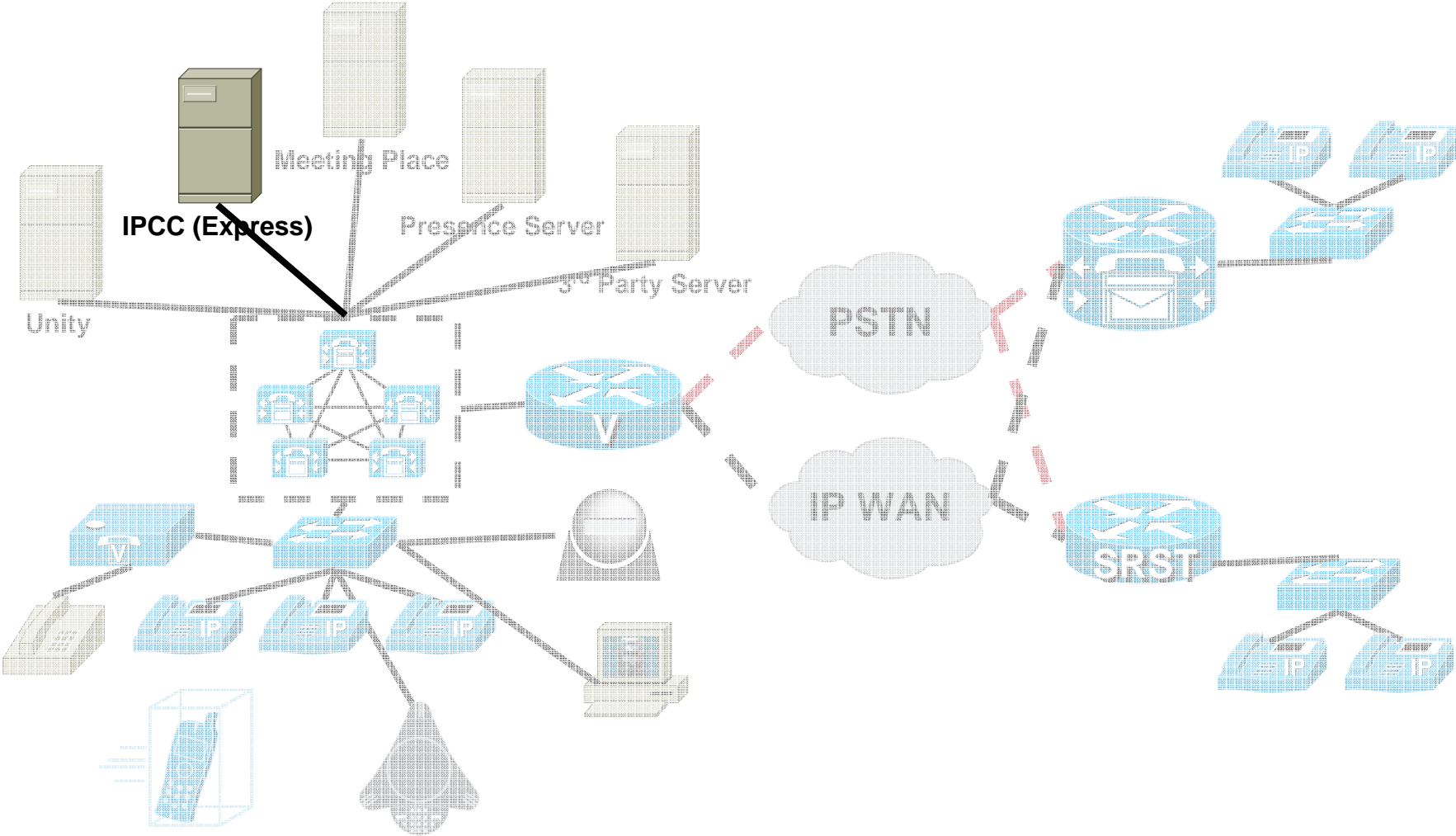


# Voice and Unified Messaging

**Depending on the deployment size and complexity different solutions are available:**

- **Cisco Unity**
  - Large scale enterprise voice and unified messaging.
  - Complex multi-site.
  - Thousands of users.
- **Cisco Unity Connection**
  - Single site enterprise voice and unified messaging
  - Up to 1500 users
- **Cisco Unity Express**
  - Router based voice messaging.
  - Small deployments.

# Cisco IP Contact Center



# Cisco Contact Center Vision

**Create an open communications platform leveraging voice and data technologies to facilitate geographic- and media-independent customer interaction**

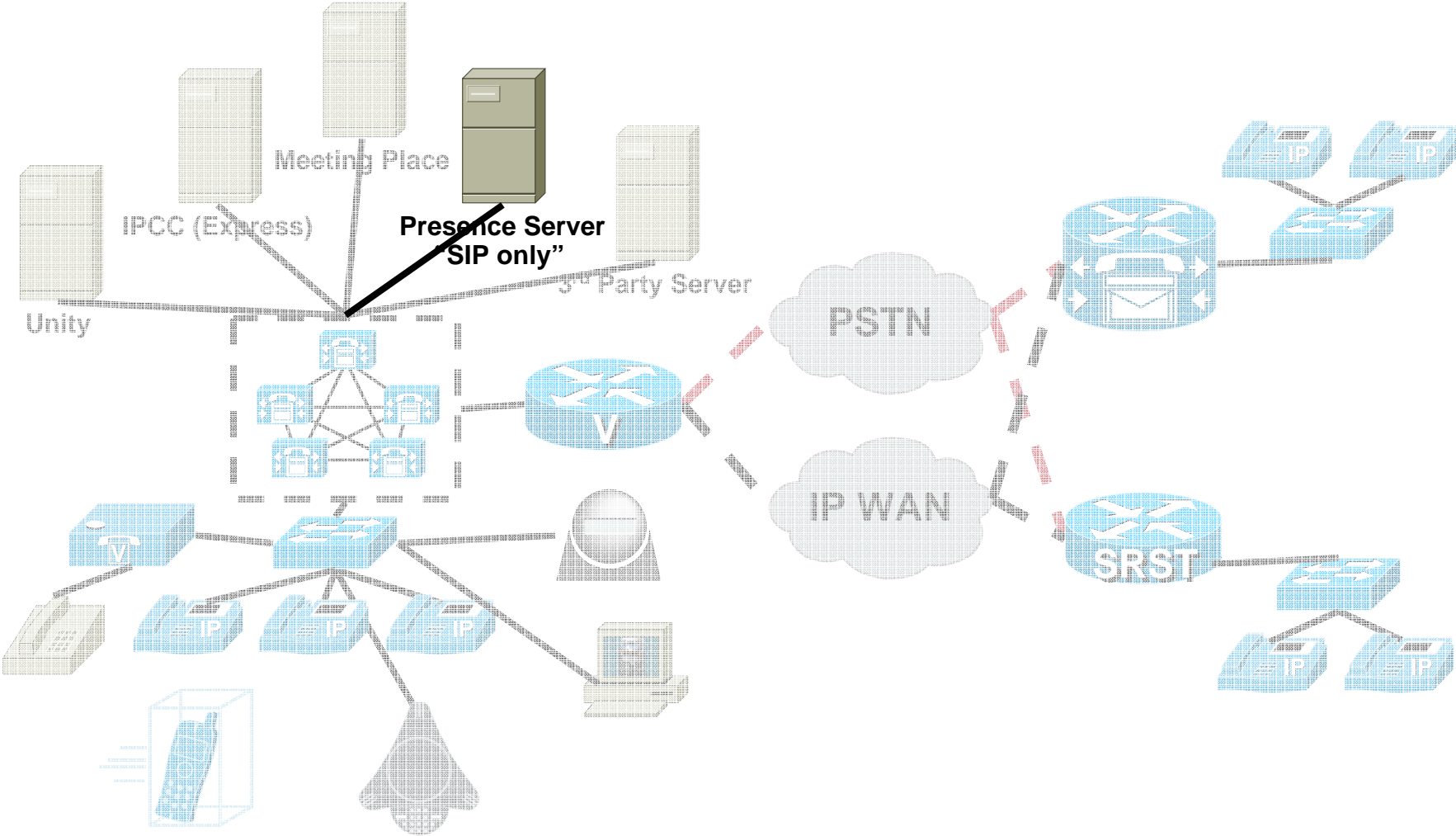


# IPCC Enterprise versus IPCC Express

Criteria	Cisco IPCC Express Edition 4.0	Cisco IPCC Enterprise Edition 7.0
Virtual Contact Center in <i>Centralized</i> Call Deployment Model (One CallManager Cluster)	✓	✓
Virtual Contact Center in <i>Distributed</i> Call Deployment Model (Multiple CallManager Clusters)	⊘	✓
Integrated Multi-Channel	⊘	✓
Blended Inbound/Outbound	⊘	✓
Scalability > IPCCX to 300 agents, IPCCE to thousands	⊘	✓
Redundancy (NOTE: IPCC Enterprise <b>has superior redundancy</b> to that provided by IPCC Express <b>e.g. preservation of calls/data in queue</b> )	partial	✓
Pure IP Solution Integrated with Cisco CallManager	✓	✓
Multi-site and heterogeneous IP and TDM support with ICM	✓	✓
Simplified installation and administration, full integration with Cisco CallManager	✓	⊘
Integrated IVR	✓	⊘



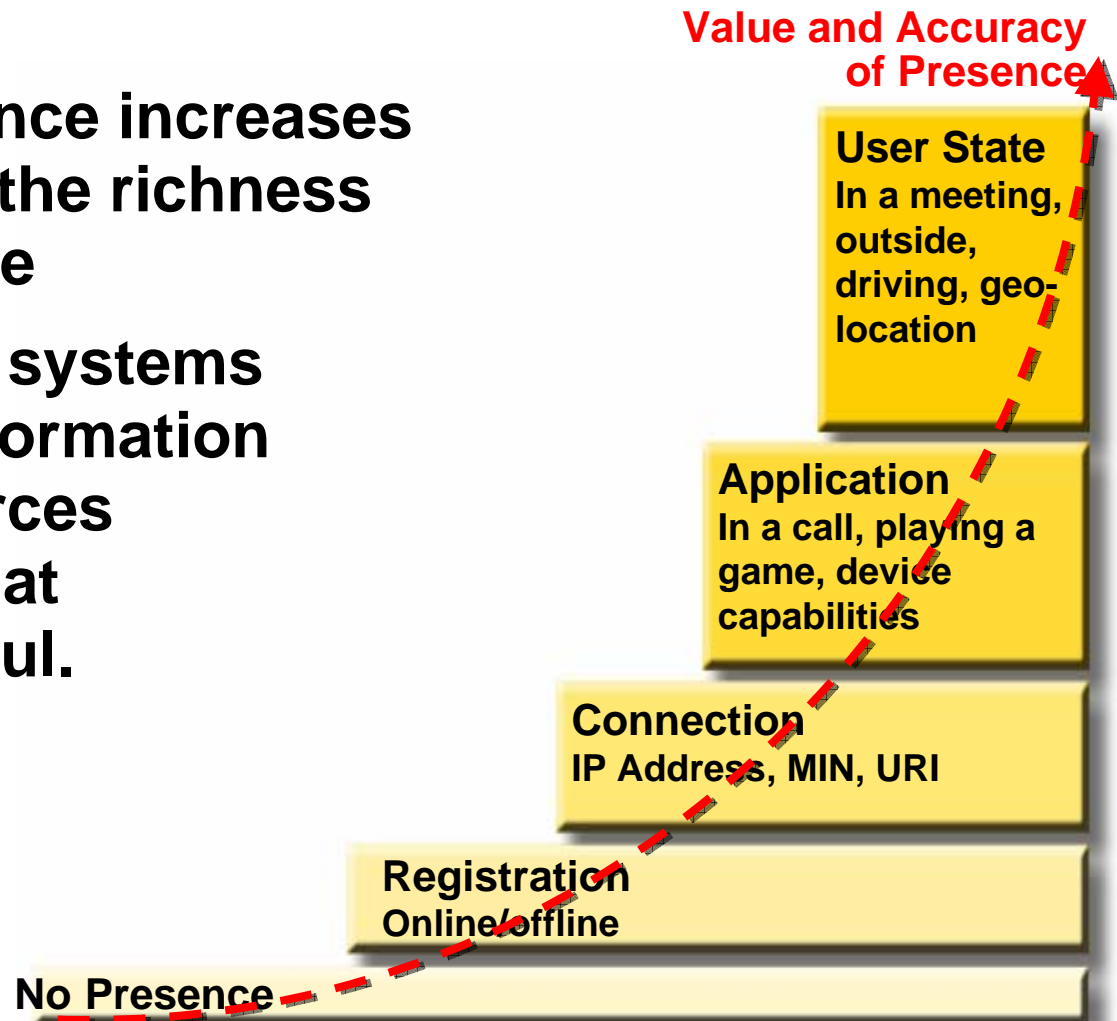
# Cisco Unified Presence Server



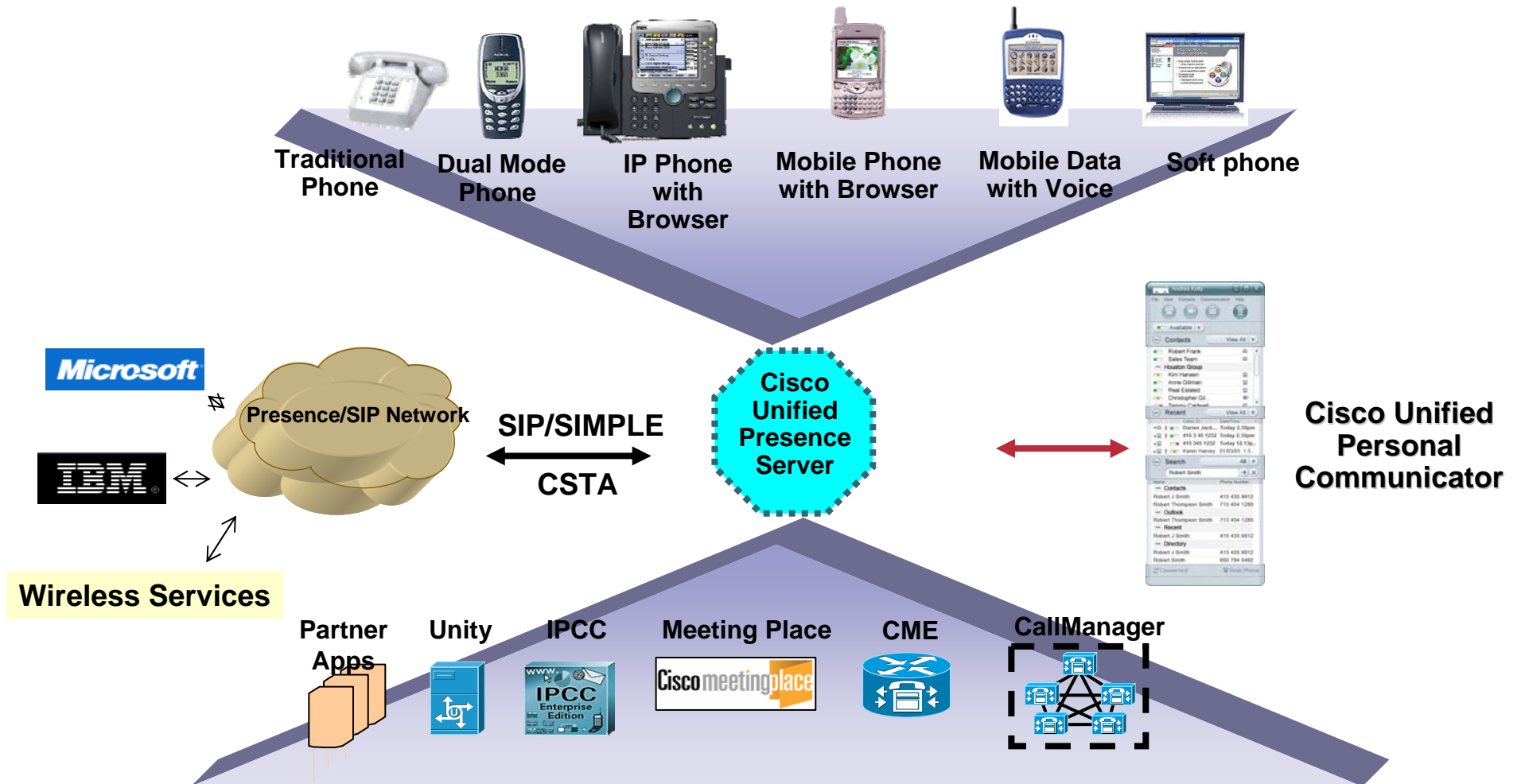
# Presence Overview: Introduction

The value of Presence increases exponentially with the richness of the data available

The best Presence systems gather accurate information from the most sources and apply policy that makes it most useful.



# Cisco Unified Presence Server: Overview



# Cisco Unified Personal Communicator

**Seamlessly integrates a wide variety of communication applications and services into a single, rich media interface, helping workers communicate more effectively.**

- Leverage reach ability and presence indicators
- Use video to exchange ideas “face-to-face”
- Escalate communication methods for more effective interactions
  - Voice, Video, Web Collaboration
- Communicate anywhere, anytime
- Pure SIP based Solution



# Cisco Unified Presence Server: IP Phone Messenger

- **Unified Client users see other user's IP phone's on/off hook states**
- **Users can send or reply to messages from their IP Phones using predefined templates or composing text messages**
- **Users can call back IM senders by hitting 1 button.**
- **Will also integrate with other IM clients and presence sources beyond UPS 1.0**



# Summary

- **Cisco has everything today's communication needs.**
- **Products in Cisco Communications build a whole Solution for converged networks. Ranging from Infrastructure to endpoints.**
- **Columbus is the follower of Golden Bridge, a wider approach that starts in the development phase.**
- **Cisco Unity is available in Voicemail only mode and in the unified messaging using one message store for all messages**
- **Contact Server Solutions from Cisco System address small to large contact center environments.**
- **With the Cisco Unified Presence Server the next step in Unified Communications is available.**

# CISCO SYSTEMS





**Understanding Cisco Enterprise IP Communication**

# **Understanding Cisco Unified CallManager Concepts**



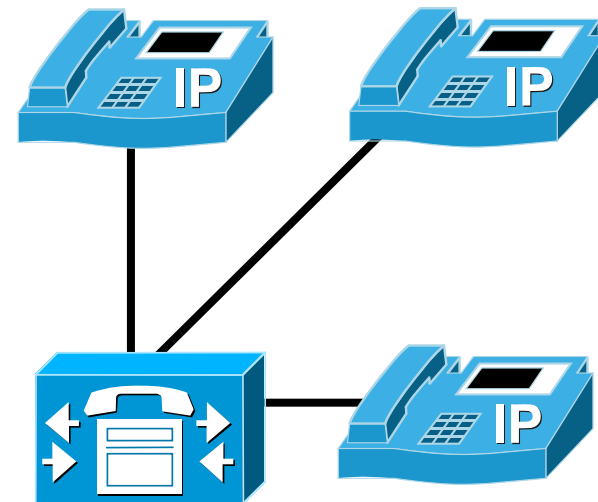
# Objectives

- **Cisco Unified CallManager Functions Overview**
- **Cisco Unified CallManager Telephony Features**
- **New features in Cisco Unified CallManager 5.0**
- **Contrast CallManager 4.x and CallManager 5.0**
- **Licensing Compliance**

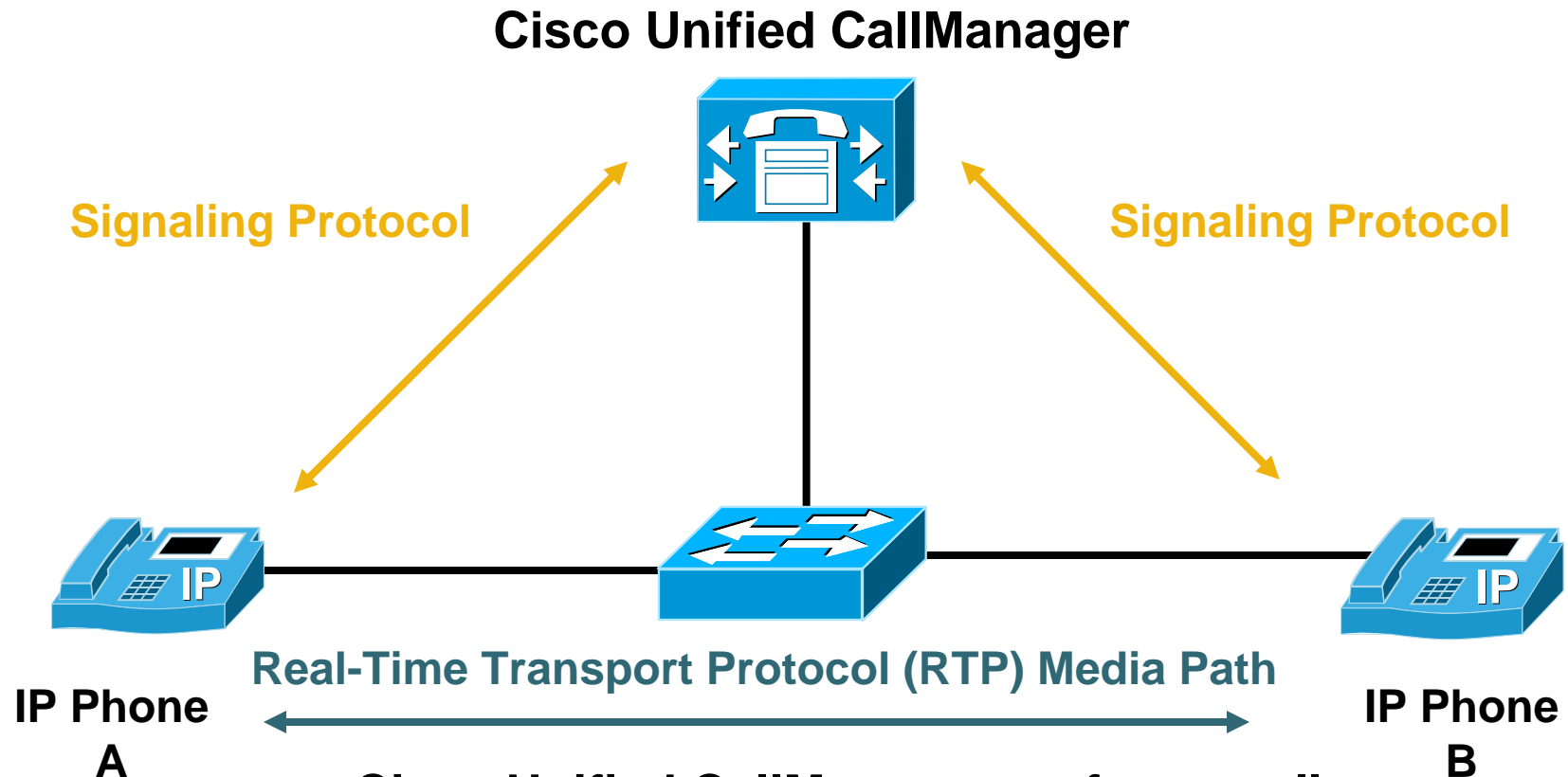
# Cisco Unified CallManager Functions Overview

## Cisco Unified CallManager is the heart of a Cisco IP Communications Environment

- Call processing
- Signaling and device control
- Dial plan administration
- Phone feature administration
- Directory services
- Programming interface to external applications



# Cisco Unified CallManager Functions (Cont.)



- **Cisco Unified CallManager performs call setup and maintenance tasks using a Signaling Protocol (Skinny/SIP).**
- **IP Phones stream audio using RTP.**

# Cisco Unified CallManager Telephony Features

- Abbreviated dial
- Barge and Conference Barge
- Call-back
- Call park and pickup
- Immediate divert to voice mail
- Multiparty conference-ad-hoc with add-on, meet-me features, Conference list and drop any party (ad-hoc conference)
- Music on hold
- Privacy
- Transfer: blind, consultative, direct transfer of two parties on a line
- Video telephony
- Web dialer-click to dial

**Note: Visit the CallManager 5.0 Data Sheet Page for a complete list of supported features:**

[http://www.cisco.com/en/US/products/sw/voicesw/ps556/products\\_data\\_sheet0900aecd8042403e.html](http://www.cisco.com/en/US/products/sw/voicesw/ps556/products_data_sheet0900aecd8042403e.html)

# New Features in Cisco Unified CallManager 5.0

## **Cisco Unified CallManager 5.0 key features include:**

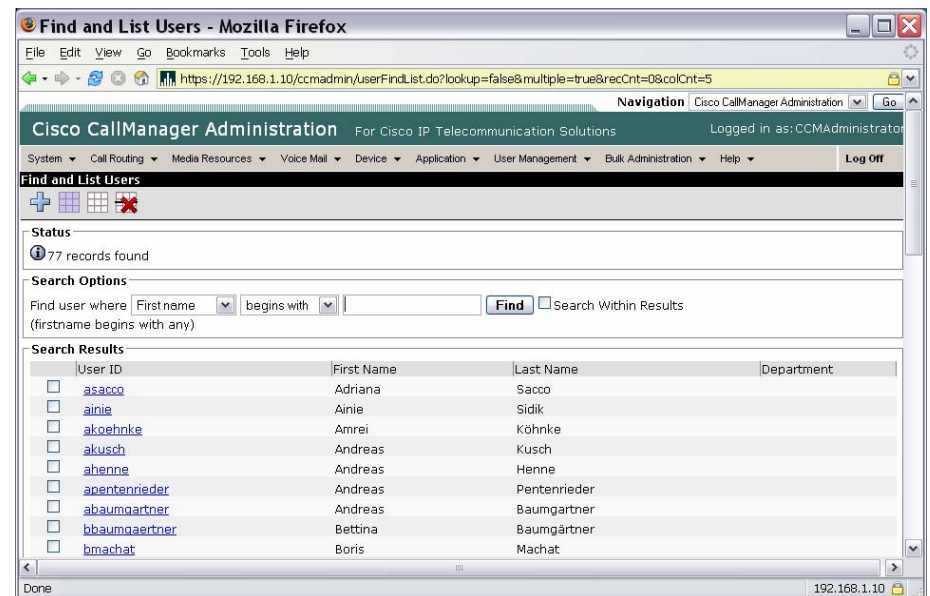
- **Appliance model**
  - Improved installation and upgrades
- **Administration enhancements**
- **Expanding SIP portfolio**
  - SIP line side
  - Enhanced networking
  - Presence with Unified Presence Server
- **Integrated Presence using Busy Line Field**
- **Japanese, Chinese, and Korean Character Set Support**
- **Licensing compliance**

# Cisco Unified CallManager Appliance Model

- **Complete hardware software solution**
- **Alternative operating system**
  - Improve installation and upgrade
  - Increased security and reliability
- **Software solution**
  - Cisco Security Agent included as with Windows version
- **Interfaces provide access to system**
  - Administration via CLI and GUI
  - Third-party access is through documented APIs

# Administration Enhancements

- Administration improvements
  - Easier and quicker user add and delete
  - Line Improvements
    - Arrange lines on 7914
  - Scheduled provisioning
    - Bulk Administration Tool
  - Navigation bar is now vertical and always present
  - Toolbar always available at top of screen
- Cisco common look and feel

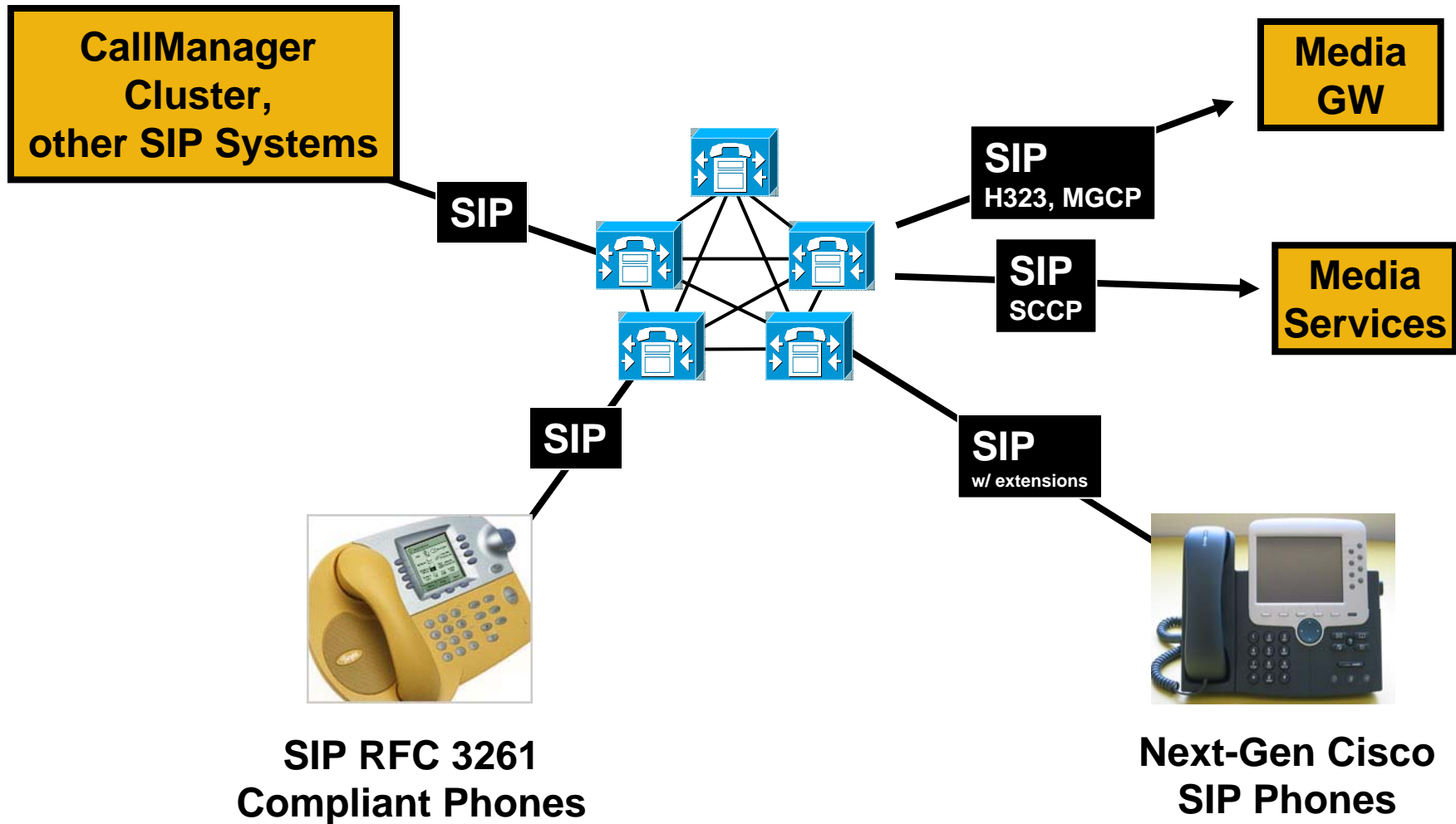


# SIP Support

- **Support of industry wide accepted SIP behaviors**
  - SIP registration
  - SIP basic call
  - SIP ‘Best Practices’ for core feature support
- **Support key Cisco Unified CallManager call control features**
  - Link into core CallManager call control
  - Support for Core CTI based applications
- **Support of third-party SIP phones**
  - Basic feature set supported.
- **Support for SIP trunks**
  - Includes intercluster trunks
- **Support for Presence**




# SIP Possibilities Overview CCM 5.0



# Cisco Unified CallManager 5.0 Presence Features

## Busy-line field is now available at CCM

- Speed-dial buttons
- Call history logs
- Directory

Icon	Description	Status
	SpeedDial	Unknown
	BLF SpeedDial	Busy
	BLF SpeedDial	Idle

# Multi-Byte Character Support (Japanese)



Incoming Call



Corporate Directory



Speed Dials



Directories

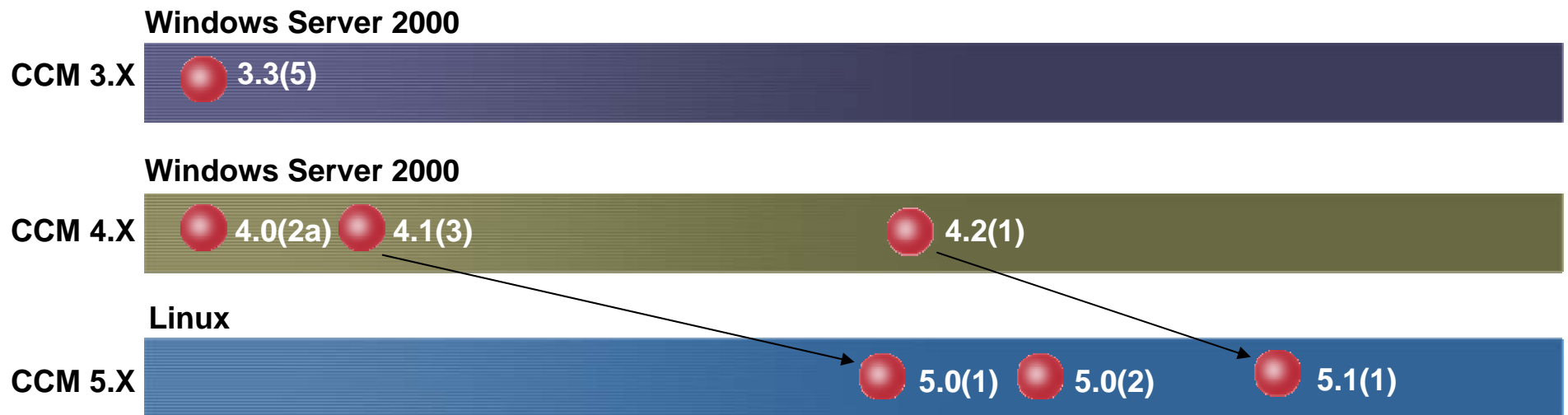
# Contrast

## Cisco Unified CallManager 4.x and 5.0

**Cisco Unified CallManager 5.0 is a new development train:**

- **Feature parity with Cisco Unified CallManager 4.1(3)**
  - 4.2 features won't be included in initial 5.0(1) releases.
- **Most 4.1(3) applications will be supported in 5.0(1)**
  - Main exception: Personal Assistant.
  - Check compatibility information for updated information.
- **4.x and 5.x train will be merged from a feature perspective.**

# Contrast Cisco Unified CallManager 4.x and 5.0



# Licensing Compliance

**Cisco Unified CallManager licensing compliance separated for Devices and Application.**

## **Device licenses:**

- **The maximum number of provisioned devices (IP phones, video devices) in CCM database will be tracked and enforced**
- **CCM will only be able to support the number of devices purchased licenses for**

## **Application licenses:**

- **The CCM software will be tied to a server via mac address**

**Licenses will be created and distributed in accordance with Cisco FlexLM process used for Cisco Unity today**

# License Unit Report

The screenshot shows the Cisco CallManager Administration interface. The page title is "License Unit Report". The status is "Ready". The "License Unit Distribution" section contains two tables:

Phone License Feature			
License Server	Units Authorized	Units Used	Units Remaining
192.168.1.10	100	11	89
<b>Total Units for Feature</b>	<b>100</b>	<b>11</b>	<b>89</b>

CCM Node License Feature			
License Server	Units Authorized	Units Used	Units Remaining
192.168.1.10	3	1	2
<b>Total Units for Feature</b>	<b>3</b>	<b>1</b>	<b>2</b>

Two callout boxes are present: "Phone Licensing" points to the "Phone License Feature" table, and "CCM Server per Cluster Licensing" points to the "CCM Node License Feature" table.

# Summary

- **Cisco Unified CallManager is the heart of a Cisco IP Communications Environment.**
- **Cisco Unified CallManager 5.0 introduces new key features, including the move to an appliance.**
- **Cisco Unified CallManager 5.0 has feature parity with Cisco Unified CallManager 4.1.3, with additional 5.0 specific features.**
- **The new licensing model also introduces license enforcement.**



# CISCO SYSTEMS





**Understanding Cisco Enterprise IP Communication**

# **Evaluating Cisco Unified CallManager Deployment Models**

# Objectives

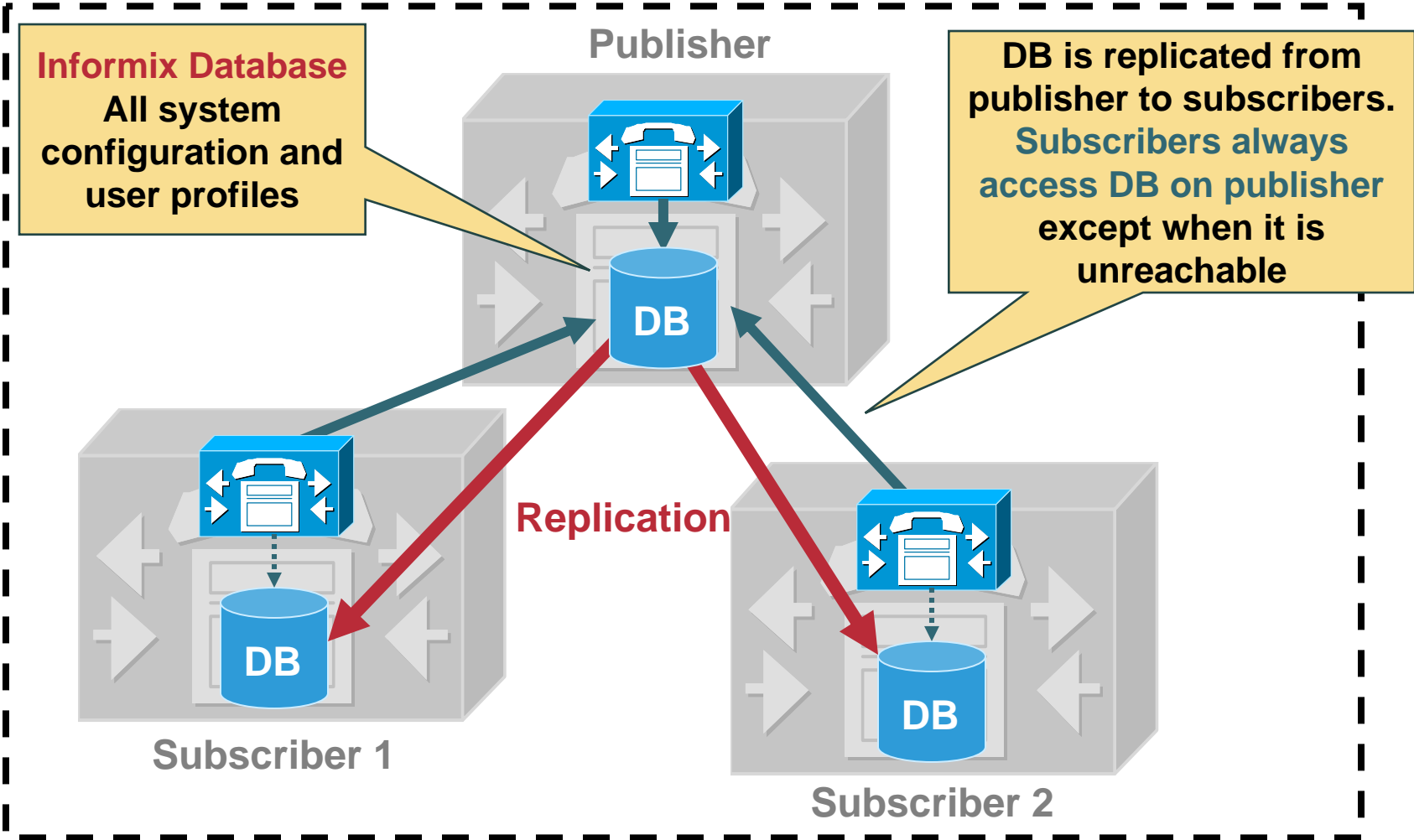
- **Cisco Unified CallManager 5.0 Redundancy**
- **Call-Processing Deployment Models**
- **Single-Site Deployment**
- **Multi site WAN with Centralized Call Processing**
- **Multi site WAN with Distributed Call Processing**
- **Clustering over the IP WAN/MAN**

# Cisco Unified CallManager 5.0 Clusters

**Clusters are used for redundancy and scalability:**

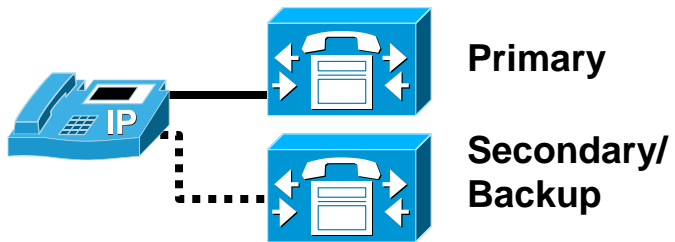
- **One Publisher maintains database.**
- **Multiple Subscribers obtain database from publisher.**
- **Up to eight servers in a Cisco Unified CallManager cluster can run call processing services.**
  - **Redundant call processing.**
  - **Shared load results in higher capacity.**
- **Role of a server in a cluster depends on activated services.**

# CallManager 5.0 Directory Architecture



**Cisco Unified CallManager Cluster**

# 1:1 Redundancy Design



- **Pro: High availability**
- **Con: Increased server count and cost**

7500 IP Phones

Cisco MCS 7845



Publisher and  
TFTP Server  
(Not Req. <1000)

Primary  
1 to 7500



Backup



15000 IP Phones

Cisco MCS 7845



Publisher and  
TFTP Server



1 to  
7500

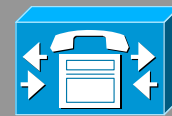
Backups



7501 to  
15000

30,000 IP Phones

Cisco MCS 7845



Publisher and  
TFTP Server



1 to  
7500

Backups



7501 to  
15000



15001 to  
22500

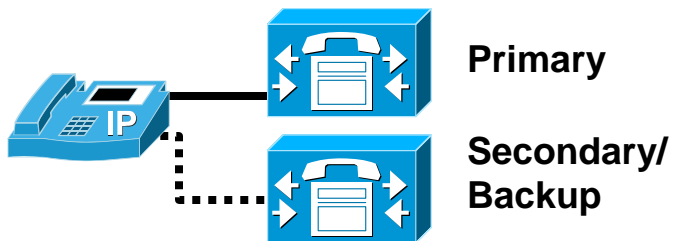
Backups



22501 to  
30000

Visit <http://www.cisco.com/go/srnd> for  
latest information.

# 2:1 Redundancy Design



- **Pro: Lower cost**
- **Con: Less redundancy**

7500 IP Phones

Cisco MCS 7845



Publisher and  
TFTP Server  
(Not Req. <1000)

Primary  
1 to 7500



Backup



15000 IP Phones

Cisco MCS 7845



Publisher and  
TFTP Server



Backups



1 to  
7500



7501 to  
15000

30,000 IP Phones

Cisco MCS 7845



Publisher and  
TFTP Server



Backups



1 to  
7500



7501 to  
15000



Backups



15001 to  
22500



22501 to  
30000

Visit <http://www.cisco.com/go/srnd> for  
latest information.

# Call-Processing Deployment Models

## Supported IP Communication deployment models:

- Single-site deployment
- Multisite WAN with centralized call processing
- Multisite WAN with distributed call processing
- Clustering over the IP WAN

## Defining characteristics of each:

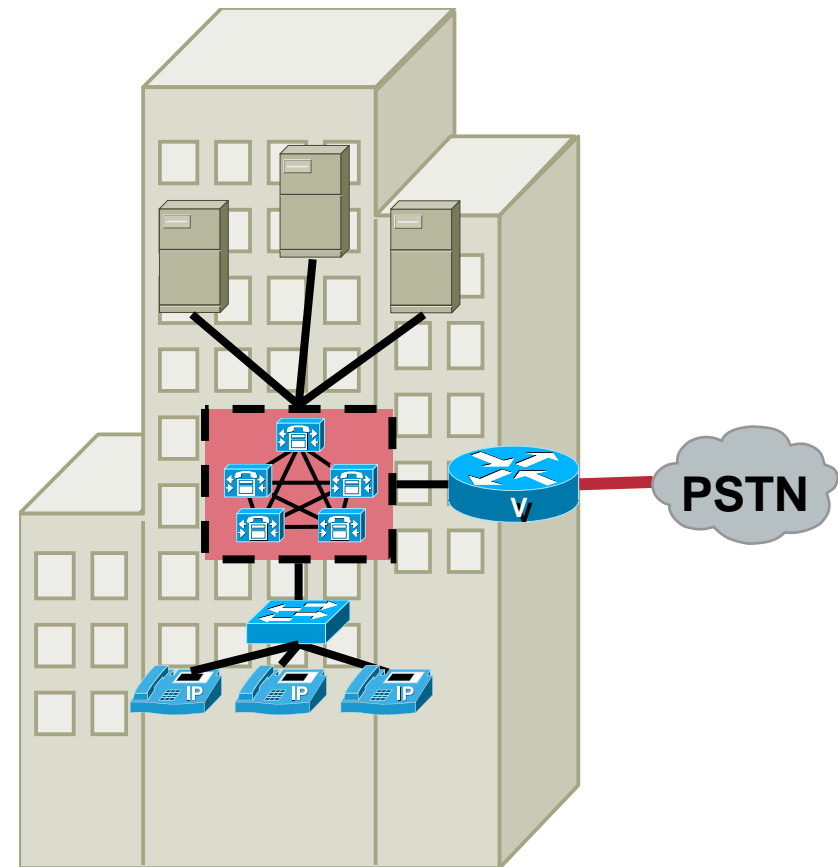
- Type of traffic that is carried over the WAN (data only or data and voice)
- Location of call-processing agent
- Size of the deployment



# Single-Site Deployment

Usually used for single site enterprise companies

- Cisco Unified CallManager servers, applications, and DSP resources at same physical location
- IP WAN (if one) used for data traffic only; PSTN used for all external calls
- Supports approximately 30,000 IP Phones per cluster

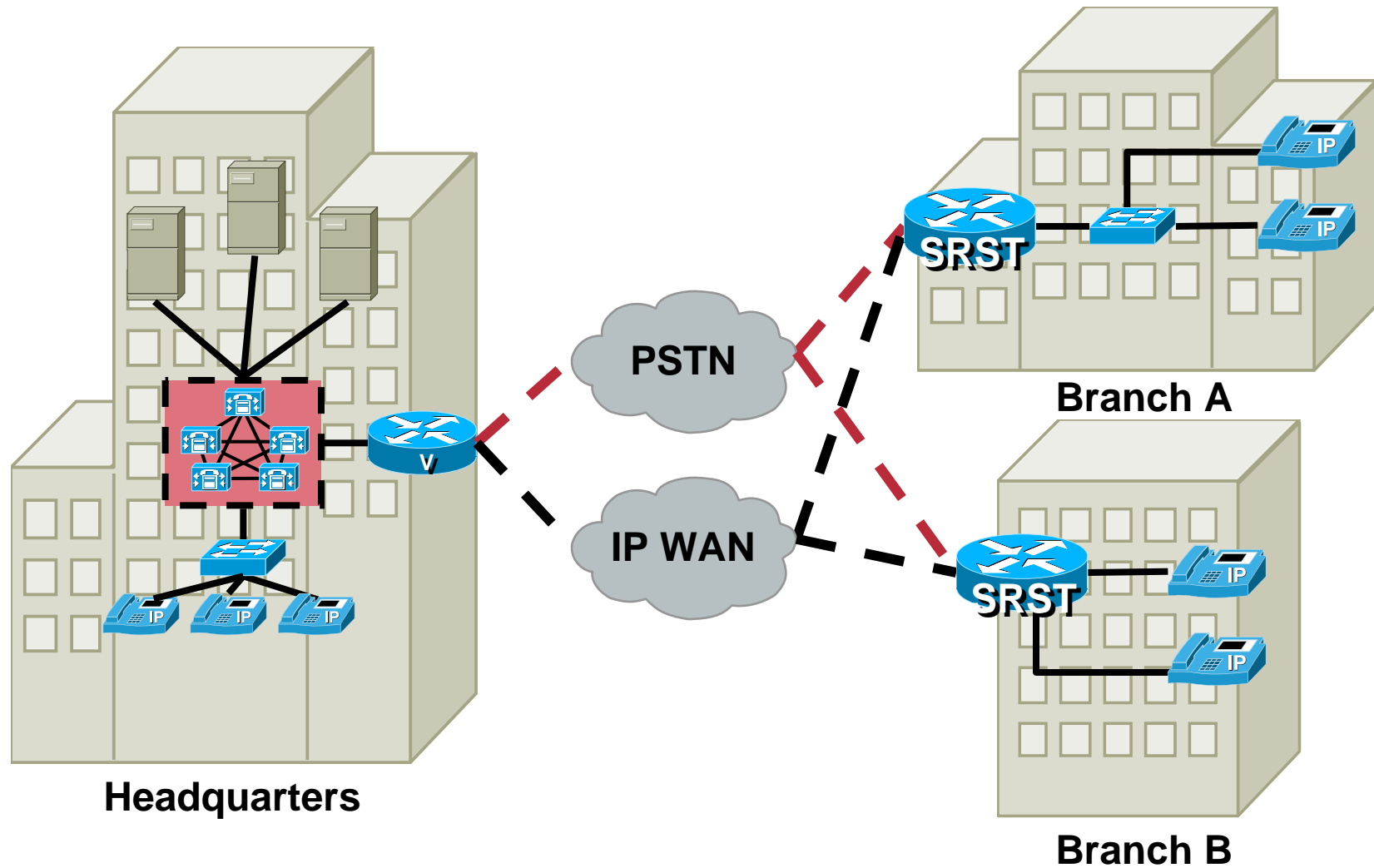


# Multi site WAN with Centralized Call Processing

**Usually used for enterprise companies with small to medium sized branch sites, centralized administration and regional proximity.**

- **Cisco Unified CallManager at central site; applications and DSP resources centralized or distributed**
- **IP WAN carries voice traffic and call control signaling**
- **Supports approximately 30,000 IP Phones per cluster**
- **Call admission control to control the number of intersite calls.**
- **Survivable remote site telephony (SRST) for remote branches**
- **Automated alternate routing (AAR) used if WAN bandwidth is exceeded**

# Multi site WAN with Centralized Call Processing

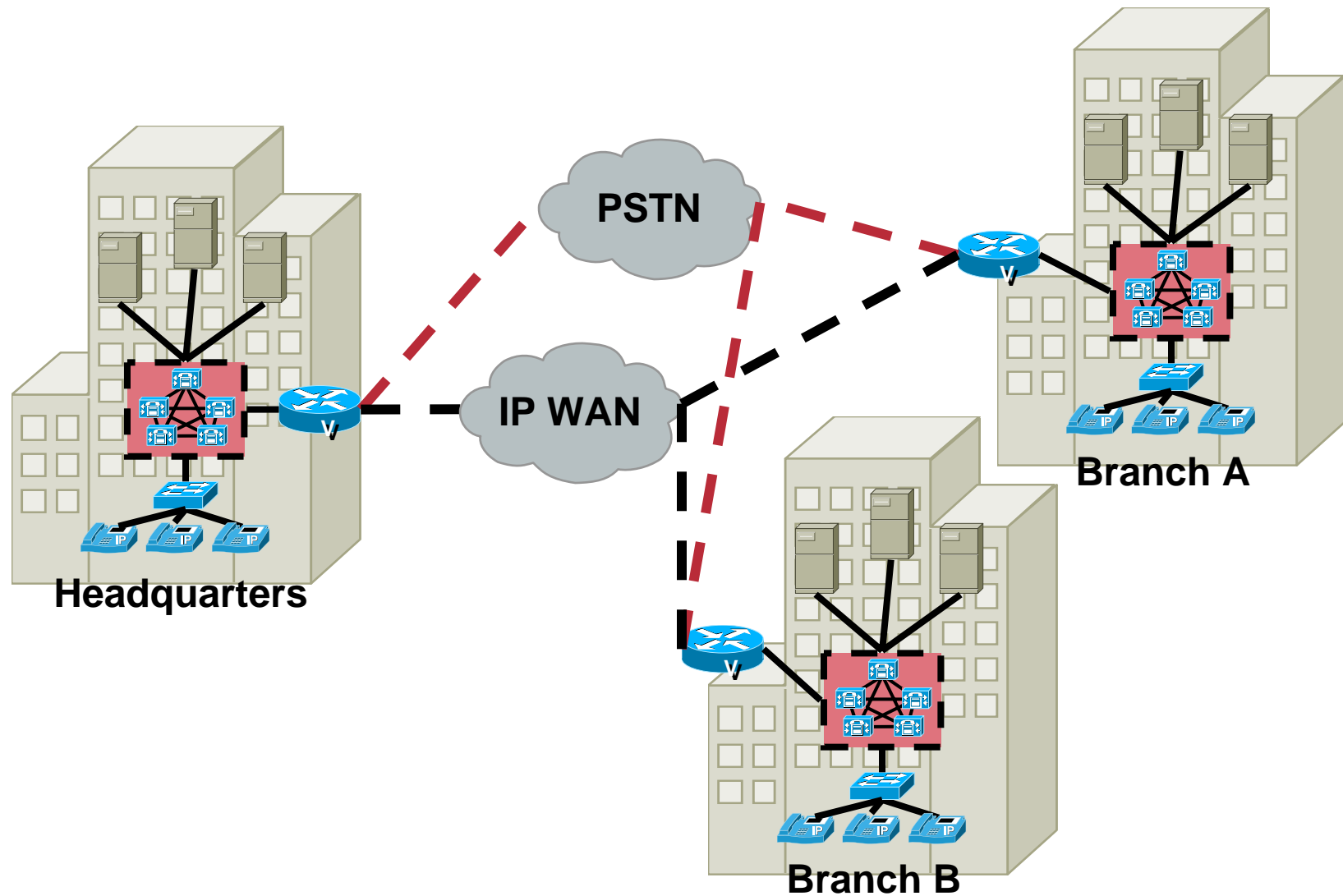


# Multi site WAN with Distributed Call Processing

**Usually used for enterprise companies with large sites in distant locations.**

- **Cisco Unified CallManager and applications located at each site.**
- **IP WAN does not carry call control signaling.**
- **Scales to hundreds of sites.**

# Multi site Distributed Call Processing: Topology

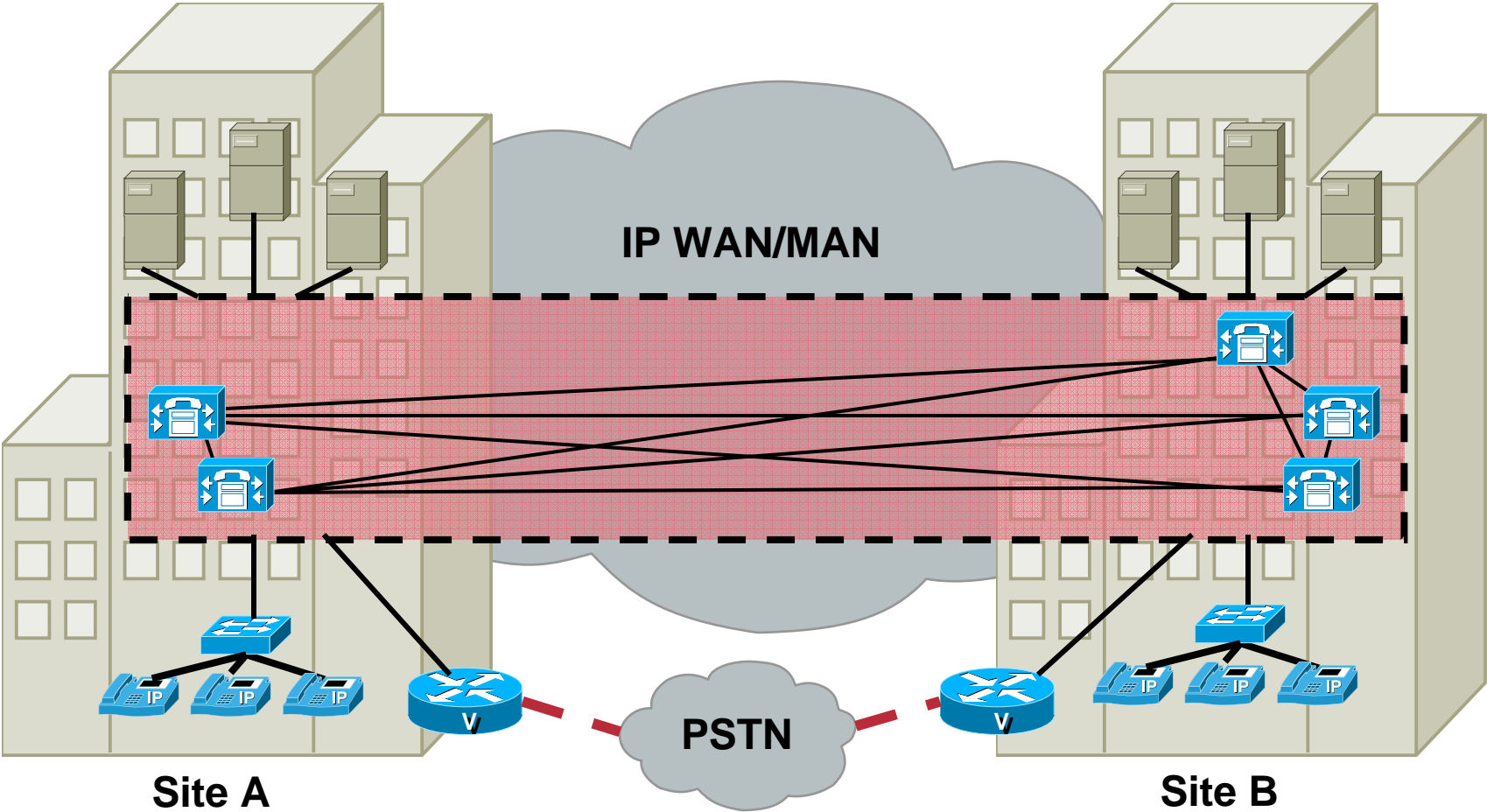


# Clustering over the IP WAN/MAN

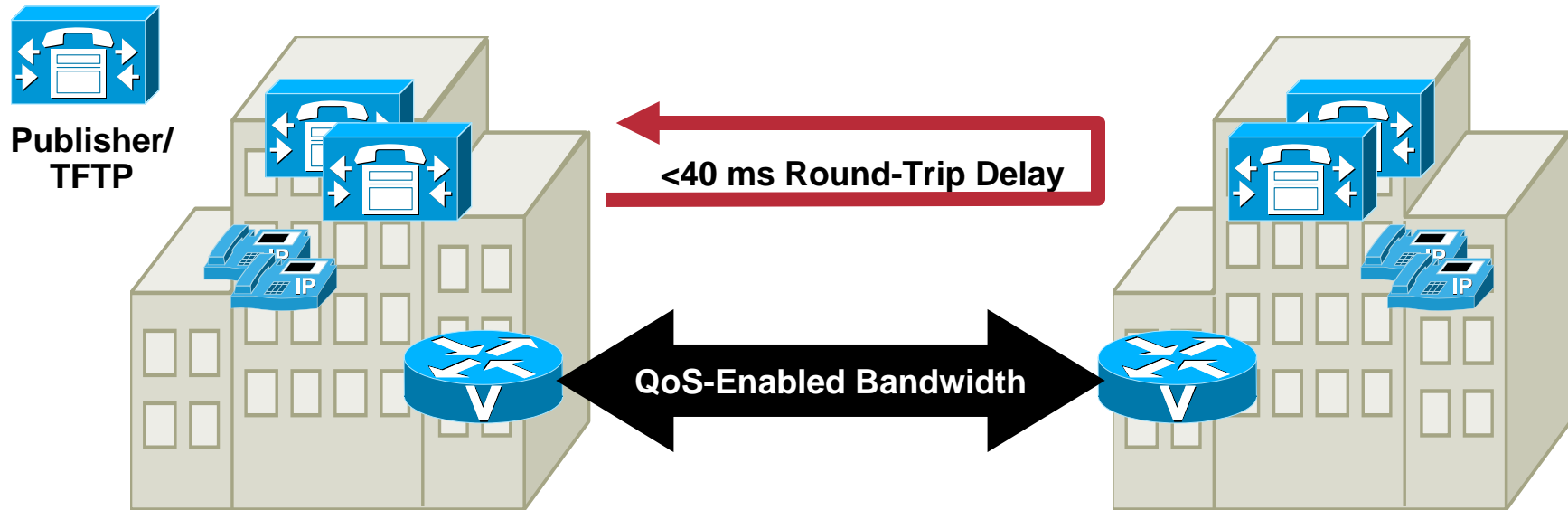
**Usually used for enterprise companies with**

- **Cisco Unified CallManager Cluster is distributed across sites.**
- **No Fallback with SRST needed.**
- **Up to eight small sites using the remote failover deployment model**
- **Failover across WAN supported (more bandwidth)**

# Clustering over the IP WAN/MAN: Topology



# Clustering over the IP WAN: Requirements with CCM 4.x

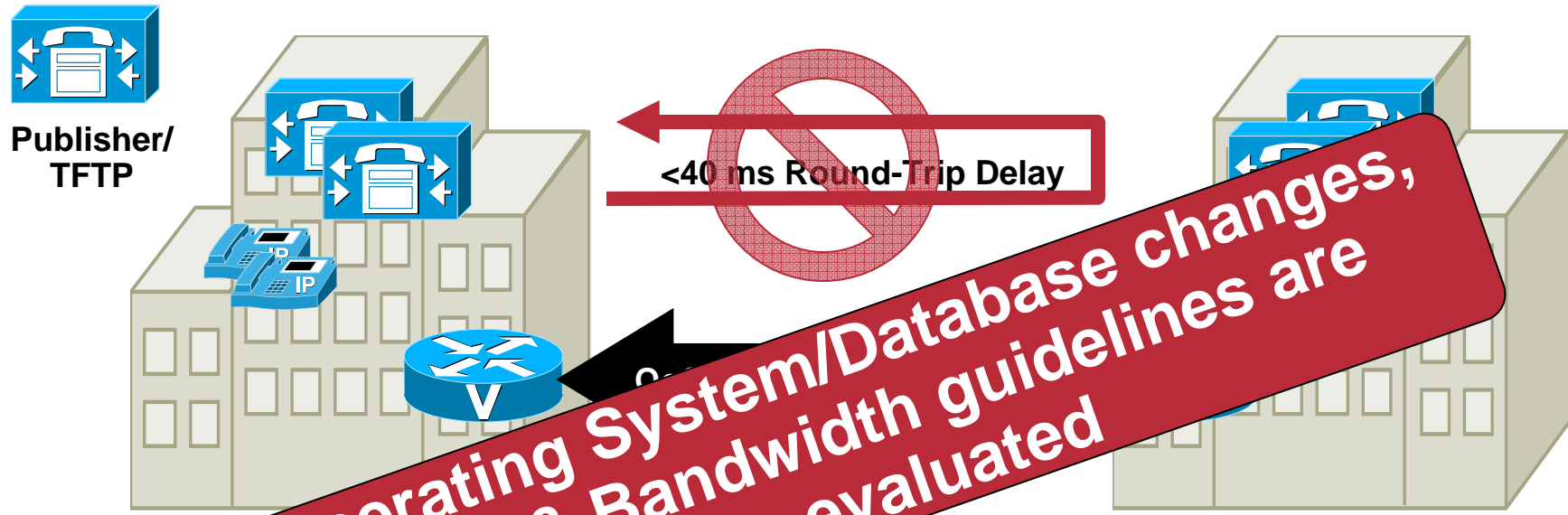


- 40-ms round-trip delay between any two Cisco Unified CallManager servers in the cluster
- 900 kbps for every 10,000 BHCAs within the cluster

Visit <http://www.cisco.com/go/srnd> for latest information.



# Clustering over the IP WAN: Requirements



- Due to changes in the Operating System/Database, RTT delay & Bandwidth guidelines are being re-evaluated
- <40 ms Round-Trip Delay between any two Cisco Unified servers in the cluster
- 900 kbps for every 10,000 BHCAs within the cluster

IP Communication SRND for CallManager 5.0 will have full details

# Summary

- **Clusters provide database redundancy. One publisher maintains the only writable database. Up to eight subscribers maintain read-only copies.**
- **There are two different failover models: the 1:1 model and the 2:1 model.**
- **Supported Cisco IP Communication deployment models are single-site, multisite with centralized call processing, multisite with distributed call processing, and clustering over the IP WAN.**

# CISCO SYSTEMS





# Understanding Cisco Enterprise IP Communication

## Evaluating Cisco IP Communication Endpoints

# Objectives

- **Endpoint Overview**
- **SCCP Phones**
- **SIP Endpoints**
- **Video Endpoints**
- **Comparing Endpoints**

# Endpoint Overview

- **A wide range of IP Communication Endpoints are available.**
- **Cisco has developed a product range to support all types of communications needs.**
- **Entry-Level phones through mid and upper level endpoints are available.**
- **3rd party vendor products are supported and can also be integrated into a Cisco IP Communications Environment.**
- **Protocols that are supported for Cisco IP Communications endpoints are SCCP, SIP and H.323.**
- **There are numerous H.323 endpoints available, but they are not covered in more depth in this course.**

# SCCP – Skinny Client Control Protocol

## SCCP Characteristic

- Fully integrated into CCM and supports all features of Cisco CallManager
- Supports all Cisco IP Phones
- This is a Cisco proprietary protocol
- Currently offers more features to Cisco IP Phones than SIP

# SCCP Phones

## SCCP- only Phones



**7902**



**7920**



**7936**



**7985**

Visit <http://www.cisco.com/go/ipphone> for latest information.



# SCCP Devices

## SCCP- only Devices



**Communicator**



**ATA-18x**



**VG248/224**

Visit <http://www.cisco.com/go/ipphone> for latest information.

# SIP – Session Initiation Protocol

## SIP Characteristics

- Is a partial implementation in CallManager 4.x and a full implementation on Cisco Unified CallManager 5.x
- The future signaling protocol for Cisco Unified CallManager
- There are Phones available with limited and advanced SIP support.
  - Only fully supported by the 7911, 7940, 7960 and 7970 phones
  - Special Firmware for IP Phones is needed
- SIP is an open standard (IETF)
- Today a feature gap between SIP and SCCP exists.

# SIP Endpoints (Basic)

## “Existing” Basic Cisco SIP Phones



**7912**

**7940/7960**

## Very Basic SIP



**Generic 3<sup>rd</sup>-Party**

Visit <http://www.cisco.com/go/ipphone> for latest information.

# SIP Endpoints (Advanced)

## “Enhanced” SIP Phones

## PC-Application



7911



7941/7961



7970/7971



Cisco Unified  
Personal  
Communicator

Visit <http://www.cisco.com/go/ipphone> for latest information.

# Video Endpoints

**As with Cisco IP Communication the three major signaling protocols for Cisco IP Video are SIP, SCCP and H.323**

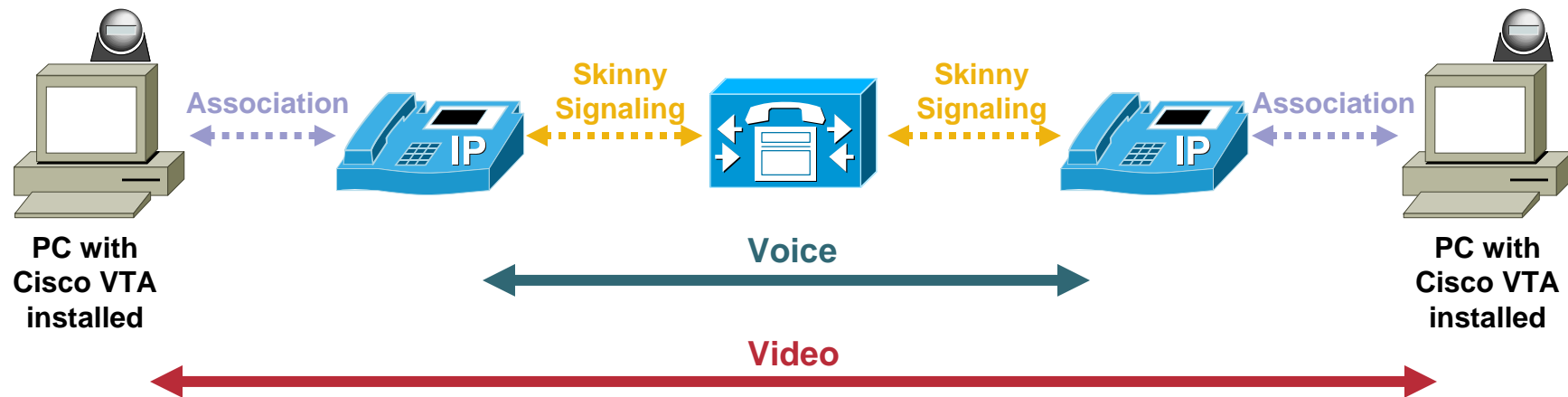
- **CCM 5.0 is the first version that supports video with SIP**
  - **SIP endpoints and trunks are now video enabled.**
  - **Currently Cisco has no SIP video endpoints.**
- **Video was first introduced in Cisco Unified CallManager version 4.0**
  - **SCCP video is still relatively new, but the relative low cost of Cisco VTA is causing rapid adoption of this type of endpoint.**
  - **H.323 video endpoints can be integrated into Cisco CallManager, examples of these devices are H.323 room systems and stand alone video terminals.**

# Cisco VT Advantage

- **Must use Cisco IP Phone 7940/7960/7970/7971**
- **PC based application**
- **Video Standards include H.263 and Cisco Wideband codec**
- **Video Resolution up to 30 fps SIF, QSIF, CIF, QCIF, VGA**
- **All phone features are maintained**



# Cisco VTA Component Interaction



- **Phones has to be enabled for video in Cisco CallManager**
- **Cisco VTA software on PC associates with IP Phone**
- **IP Phone registers as a video capable phone**
- **Audio on the IP Phone**
- **Video on the PC**

# Cisco 7985 Video Phone

- **Must use Skinny Call Control Protocol (SCCP)**
- **Video Standards are H264, H.263+, H263, H261**
- **Video Resolution include CIF, SIF, QCIF**
- **Audio Codecs available are G711, G722, G729**
- **Directory Search available**
- **SRST fallback, CDP available**



Visit <http://www.cisco.com/go/ipphone> for latest information.



# Tandberg SCCP Video Terminals

- Tandberg 550MXP, 770MXP, T1000MXP, 1500MXP, 2000MXP
- Video Standards are H.261, H.263, H.263+, H.264
- Video Resolution include CIF, QCIF, SIF
- Audio Codecs available are G.711, G.722, G.729AB
- XML-Services, Call Back
- No SRST fallback, No CDP



# SONY SCCP Video Terminals

- **Sony PCS TL-50 and PCS-1**
- **Video Standards**  
H.261, H.263, H.264 (CCM 4.1)
- **Video Resolution**  
CIF, QCIF
- **Audio Codecs**  
G.729a, G.728, G.723.1, G.711, G.722
- **Directory Search available**
- **No SRST fallback, No CDP**



# H.323 Video Terminals

**H.323 Video Terminals are widely available in the field.**

- **H.323 Video Endpoints are configured as H.323 Clients**
- **H.323 MCU 's are configured as H.323 Gateways**
- **You may also register H.323 Terminals on a Gatekeeper (GK)**
  - **If they are only registered at the GK then a trunk from Cisco Unified CallManager to the GK is used to route calls**
  - **If they are registered at GK *and* configured in Cisco Unified CallManager then E.164 Numbers are used with a RAS-Aggregation Trunk between Cisco Unified CallManager and GK to dynamically resolve the IP addresses.**
- **Examples of well known vendors are Polycom, Sony, Tandberg and Microsoft.**

# Video Codec Support Cisco Unified CallManager 5.0

	H.261	H.263	H.263+	H.263++	H.264	Cisco VTA Wideband
SCCP	X	X	X	X	X	X
H.323	X	X	X	X	X	
SIP	X	X	X	X	X	
H.323 ICT	X	X	X	X	X	X

# Comparing Endpoints

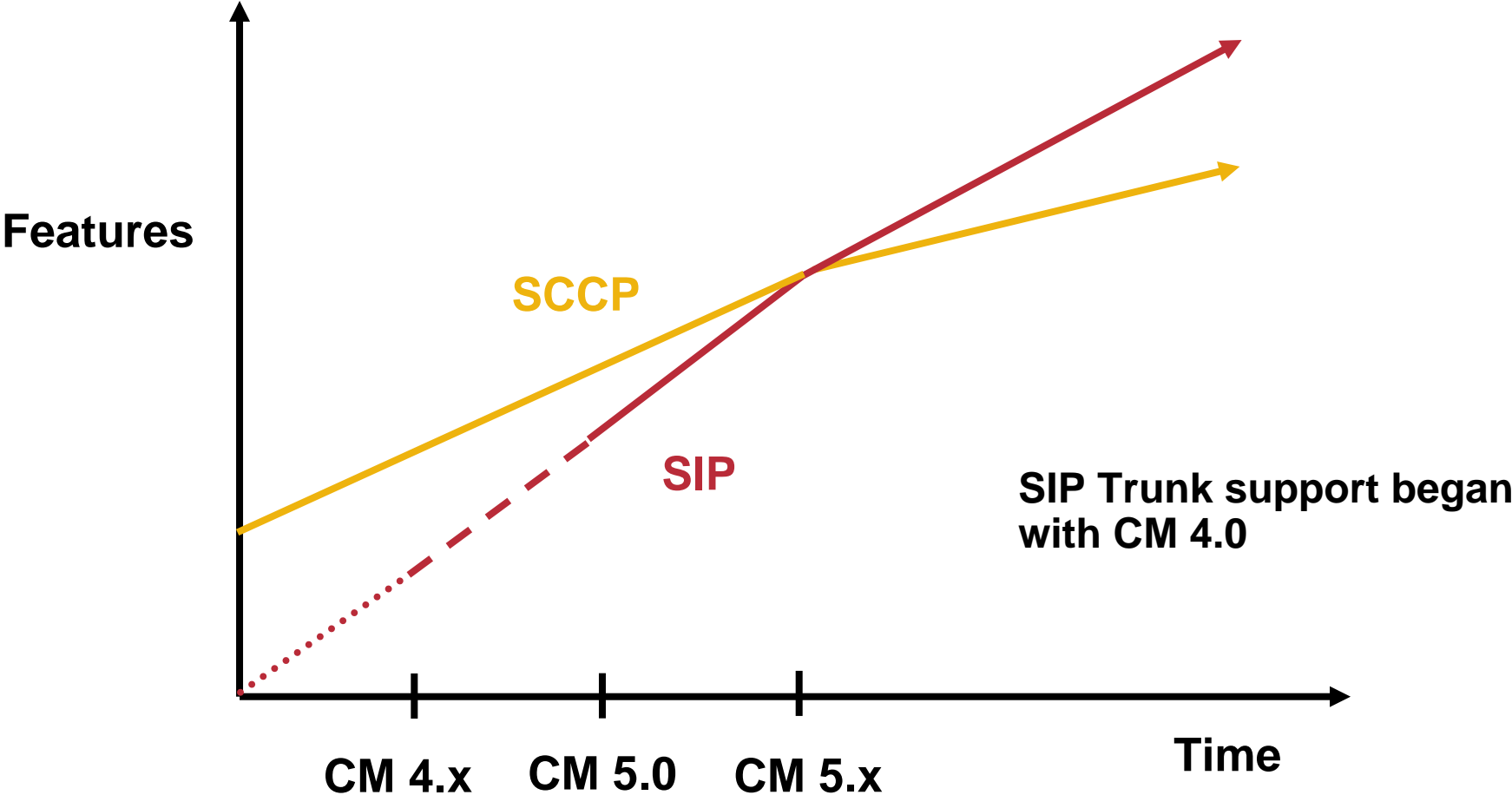
**In Cisco IP Communication environments SCCP and SIP are the major Signaling Protocols.**

**SIP is the rising star in the Cisco IP Communication signaling protocol family**

- **Service Providers will be able to offer new services to end customers**
- **SIP for Cisco Unified CallManager will offer more features than SCCP in the near future**

**H.323 is more or less a survivor from the beginning of IP Communication, but it still very relevant in relation to gateways.**

# Feature Richness SCCP - SIP



# SIP vs SCCP configuration differences

SIP	SCCP
<ul style="list-style-type: none"><li>• Configuration file via TFTP</li><li>• Softkey File via TFTP (Enhanced IP Phones only)</li></ul>	<ul style="list-style-type: none"><li>• Configuration via TFTP</li><li>• Softkey template via SCCP messages</li></ul>
Can have local dial plan	Does not have local dial plan
Registers via UDP, TCP or TLS	Registers via TCP or TLS

# Summary

- **With CallManager 5.x all available standards based IP-Phones are supported.**
- **All Cisco IP Phones support SCCP firmware**
- **The new IP-Phone models will support SIP firmware.**
- **A broad range of video endpoints are available for Cisco IP Communications**
- **SIP is the up and coming new standard for endpoints, but its features are still not up to par with SCCP**



# CISCO SYSTEMS





## Deploying a Cisco Unified CallManager 5.0 Cluster

# Cisco Unified CallManager 5.0 Architecture

# Objectives

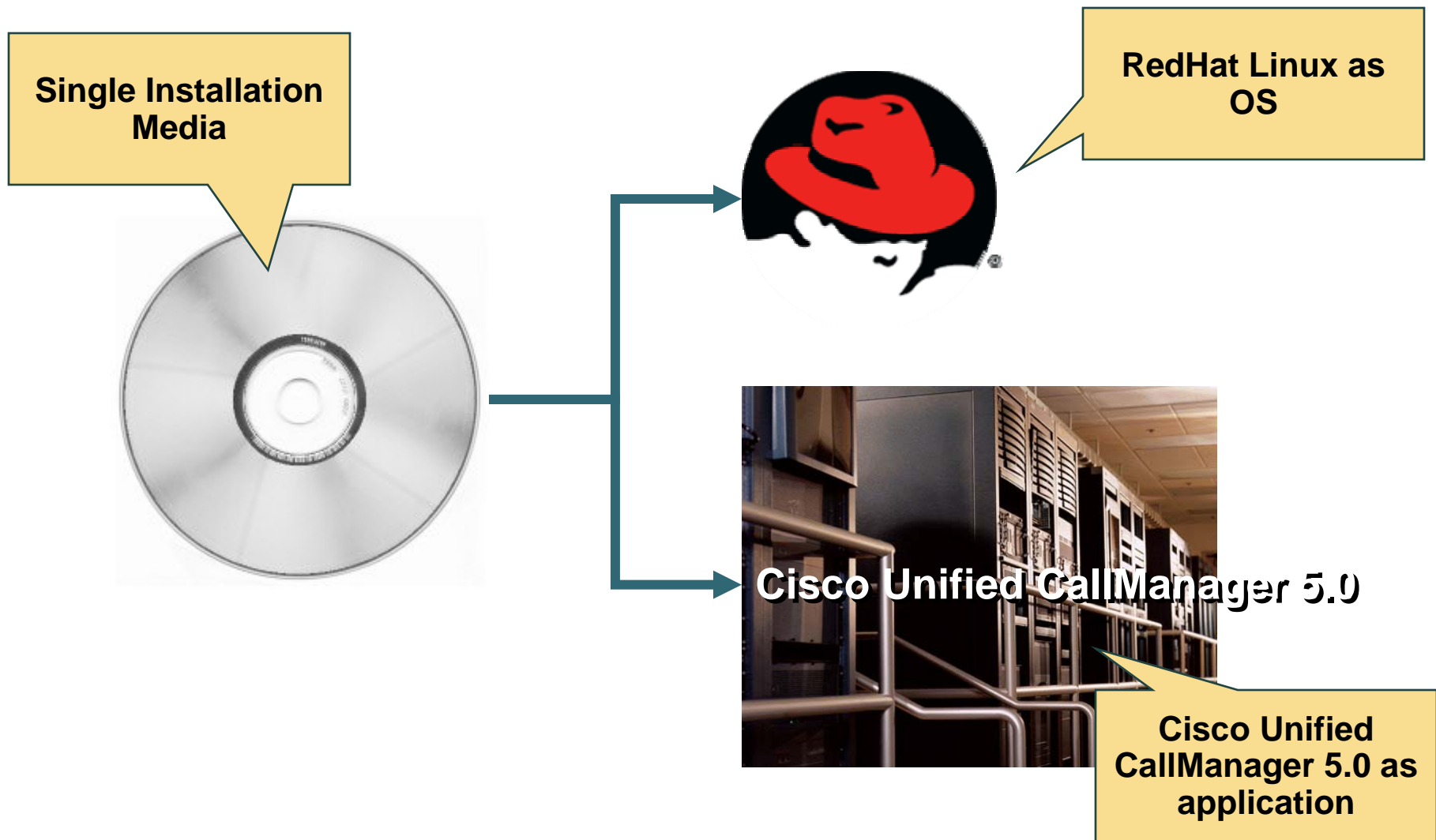
- **Cisco Unified CallManager 5.0 Overview**
- **Appliance Model**
- **Database Changes**
- **Active and Standby partitions**
- **CLI of Cisco Unified CallManager 5.0**

# Cisco Unified CallManager 5.0 Overview

## **Cisco Unified CallManager 5.0 Characteristics:**

- **Appliance model based on Linux.**
- **Combines OS and Application installation.**
- **Initial feature parity with Cisco Unified CallManager 4.1(3).**

# Cisco Unified CallManager 5.0 Overview



# Appliance Model

## **Cisco Unified CallManager 5.0 introduces the appliance approach:**

- **Avoids contact with the actual OS.**
- **User should only access the appliance via Cisco controlled tools and interfaces.**
  - **Less prone to unexpected software behavior.**
  - **Easier to test and deploy.**
- **No OS knowledge needed to operate Cisco Unified CallManager 5.0**
- **The Appliance Model uses a distinctly separate, CLI based, administration platform than CallManager.**

# Appliance Model

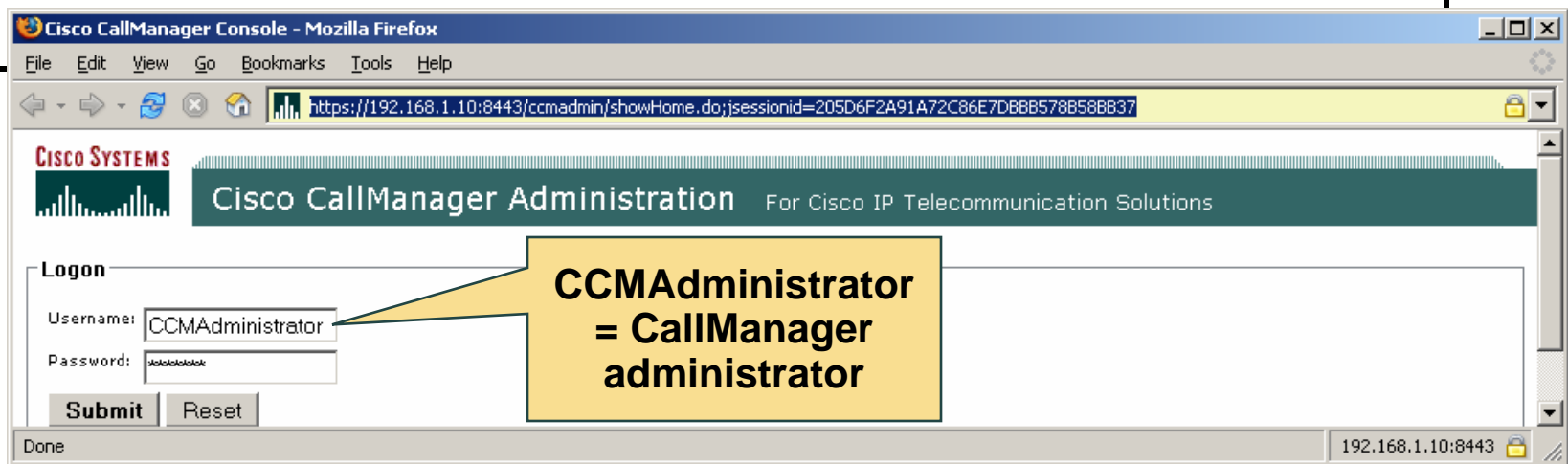
```
login as: administrator
administrator@192.168.1.10's password:

Welcome to the Platform Command Line Interface (version 1.1)

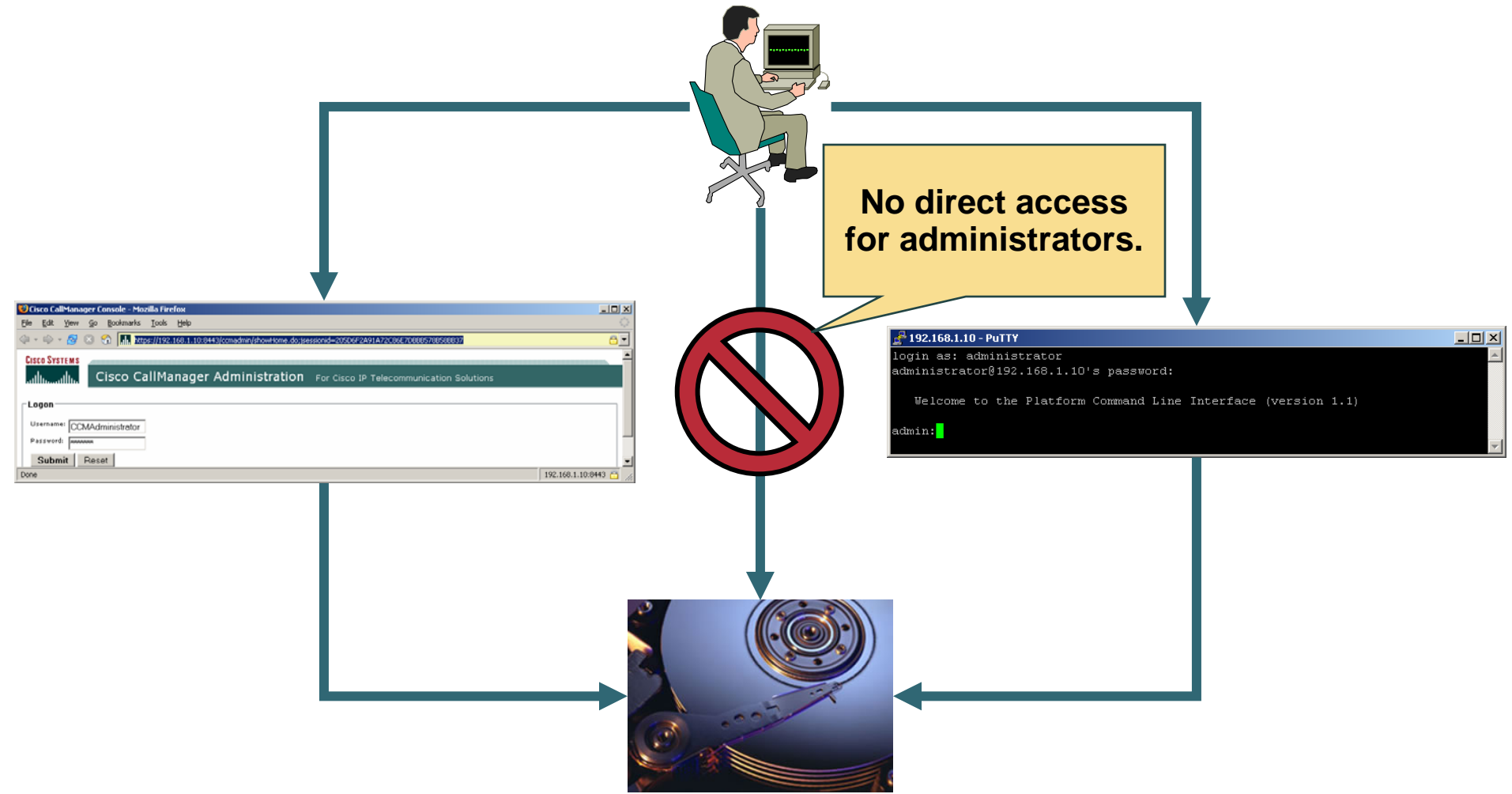
admin:
  show*
  set*
  delete*
  unset*
  file*
  utils*
  run*
```

**Administrator =  
Platform  
administrator**

admin:



# Appliance Model





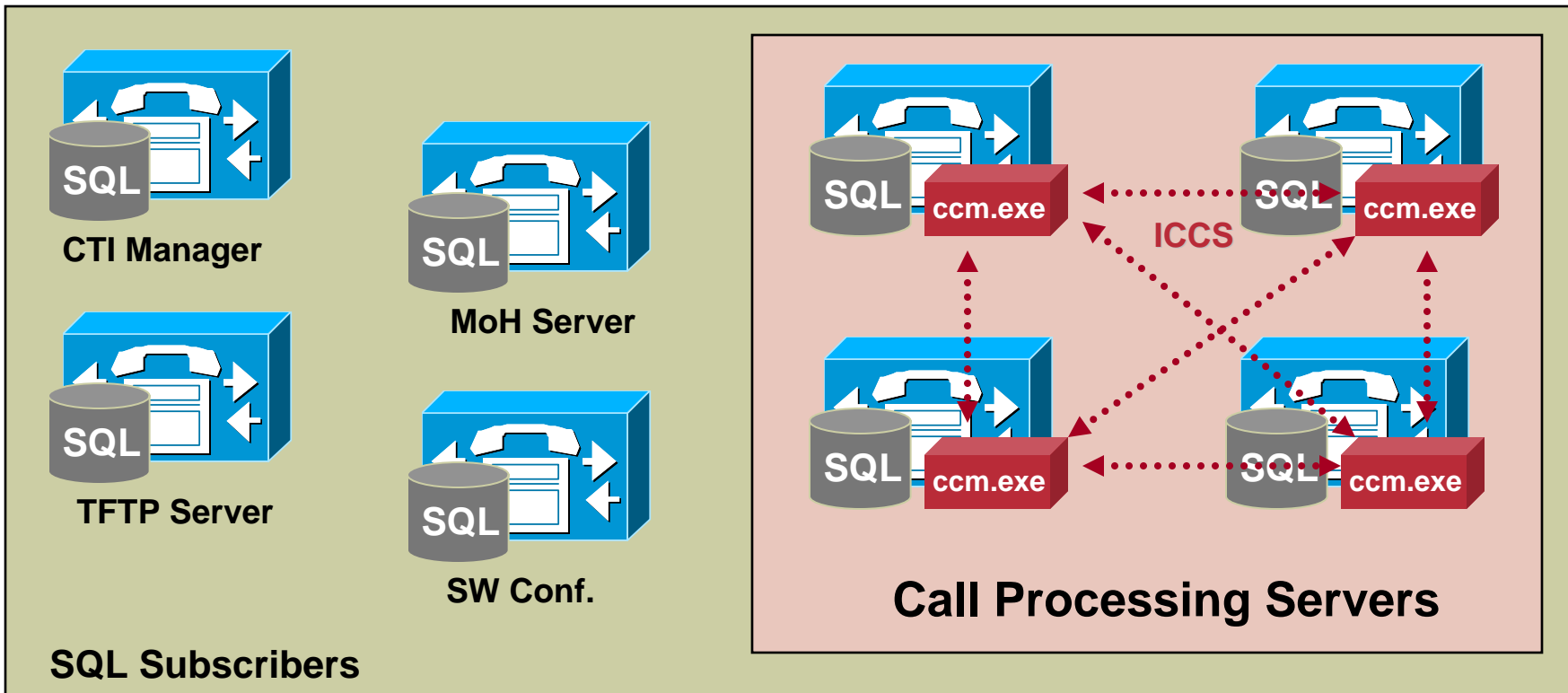
# Database Changes

**Cisco Unified CallManager 5.0 introduces a new database:**

- **Windows CallManager uses MS SQL DB.**
- **Cisco Unified CallManager 5.0 uses an IBM Informix DB.**
- **Replication mechanism is similar:**
  - **Currently no multi-master database.**
  - **Publisher still single point of failure.**
  - **May change in the future.**
- **Active and Standby partitions allow easy fallback.**

# Database Changes

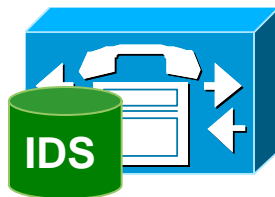
## CallManager 4.X Cluster



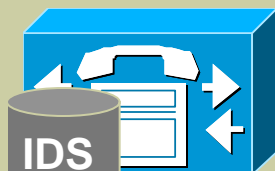
# Database Changes

## CallManager 5.0 Cluster

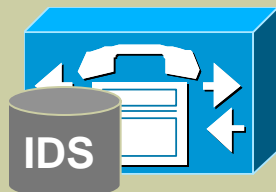
Informix Dynamic Server (IDS) Replication



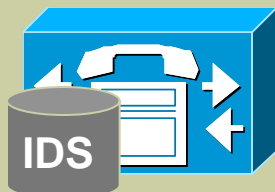
Publisher



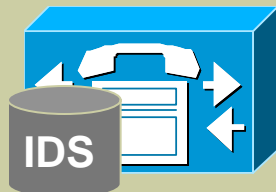
CTI Manager



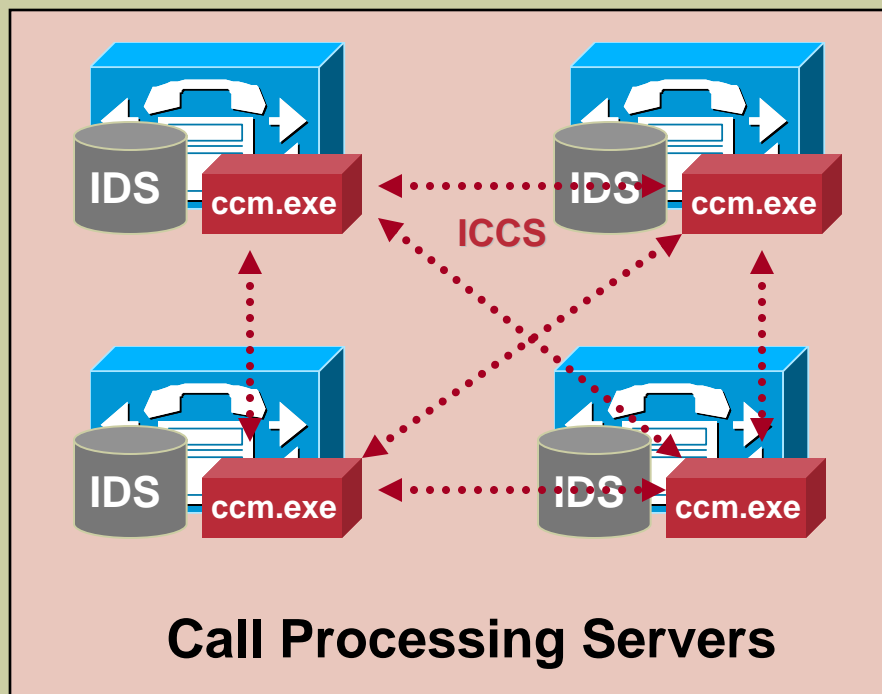
MoH Server



TFTP Server



SW Conf.



Call Processing Servers

Informix Dynamic Server (IDS) Subscribers

# CLI of Cisco Unified CallManager 5.0

## CLI used to perform platform administration:

- Direct access via console.
- SSH access via SSH Clients.
- Platform administration includes:
  - Network configuration
  - Network security configuration
  - User management for platform administration
- Similar to IOS
  - Use '?' for context sensitive help

# CLI of Cisco Unified CallManager 5.0

```
Welcome to the Platform Command Line Interface (version 1.1)
```

```
admin:
```

```
  show*  
  set*  
  delete*  
  unset*  
  file*  
  utils*  
  run*
```

```
admin:set ?
```

```
  set timezone  
  set web-security  
  set smtp  
  set account  
  set output*  
  set logging*  
  set workingdir*  
  set network*  
  set password*  
  set ipsec*  
  set cert*  
  set trace*
```

# CLI of Cisco Unified CallManager 5.0

```
Welcome to the Platform Command Line Interface (version 1.1)
```

```
admin:ls
```

```
Executed command unsuccessfully  
No valid command entered
```

```
admin:mkdir
```

```
Executed command unsuccessfully  
No valid command entered
```

```
admin:
```

**No "Linux" root  
access and  
commands.**

# Summary

- **Cisco Unified CallManager 5.0 is based on a Linux appliance.**
- **The appliance separates application and OS access.**
- **Cisco Unified CallManager 5.0 uses an IBM Informix database.**
- **The CLI can be used to perform basic platform administration in Cisco Unified CallManager 5.0.**

# CISCO SYSTEMS







**Deploying a Cisco Unified CallManager 5.0 Cluster**

# **Installing Cisco Unified CallManager 5.0**

# Objectives

- **Hardware Requirements**
- **Installation Requirements**
- **Installation Steps**
- **Calculate Licenses**
- **Install License File**

# Hardware Requirements

## **CCM5.0 has strict hardware requirements:**

- **Only supports specific hardware models.**
- **Installer script performs various hardware checks:**
  - **Platform vendor and product ID**
  - **BIOS version and release**
  - **HD raid arrays and disk health**
  - **Number of CPUs**
  - **Available RAM**
- **Installation only proceeds if ALL requirements are met.**

# Hardware Requirements (cont.)

## Hardware Requirements:

MCS 7845	Depends
MCS 7835	Depends
MCS 7825	Depends
MCS 7815	Depends



CPU	=> 2 GHz
RAM	=> 2 GB
HD	=> 72 GB

**HDs and RAM can easily be upgraded.**

# Installation Requirements

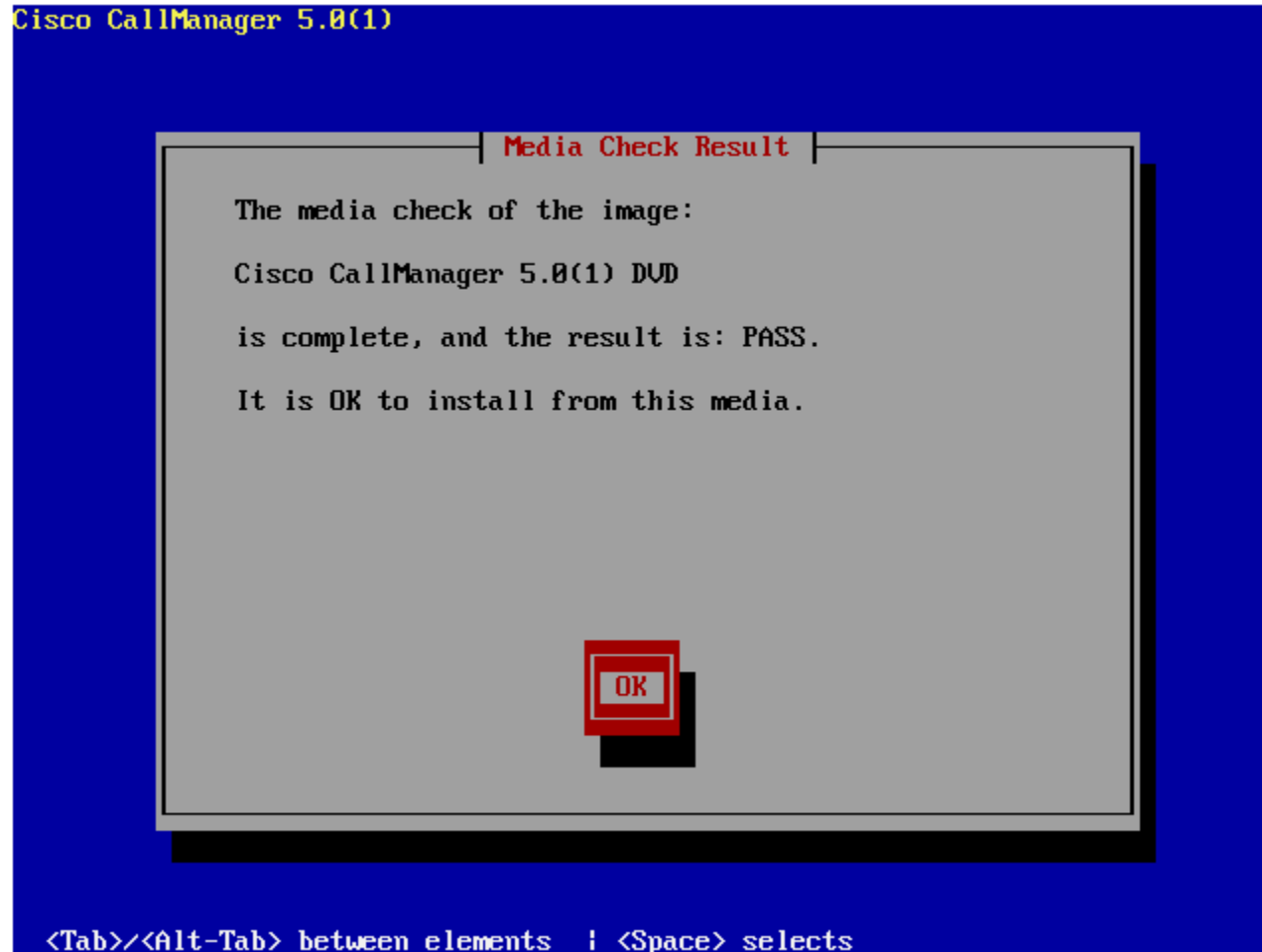
The following information is required for installation:

Functionality	Evaluation
License	Mandatory
IP Address and Default Gateway	Mandatory
Hostname and Domain	Mandatory
NTP	Highly Recommended
DNS	Highly Recommended, especially for SIP
External DHCP	Highly Recommended

# Installation Steps

- 1. Basic Installation Procedure**
- 2. Configure Timezones**
- 3. Configure static IP address**
- 4. Configure DNS Client**
- 5. Configure Platform Administrator Login**
- 6. Configure certificates**
- 7. Configure NTP**
- 8. Configure Database Access Security**
- 9. Configure SMTP**
- 10. Configure password for CCMAAdministrator**

# Basic Installation Procedure



# Basic Installation Procedure

Cisco CallManager 5.0.1.51-414

## Platform Installation Wizard

This Wizard will assist you in setting up the initial configuration of the node.

Use <Tab> and <Alt><Tab> to navigate between text entry fields and buttons in the form. Use <Space> or <Enter> to press a button. Pressing <Ok>, <Help>, <Add>, or <Delete> performs the action.

Press <Proceed> to continue, <Skip> to configure later, or <Cancel> to cancel the installation.

Proceed

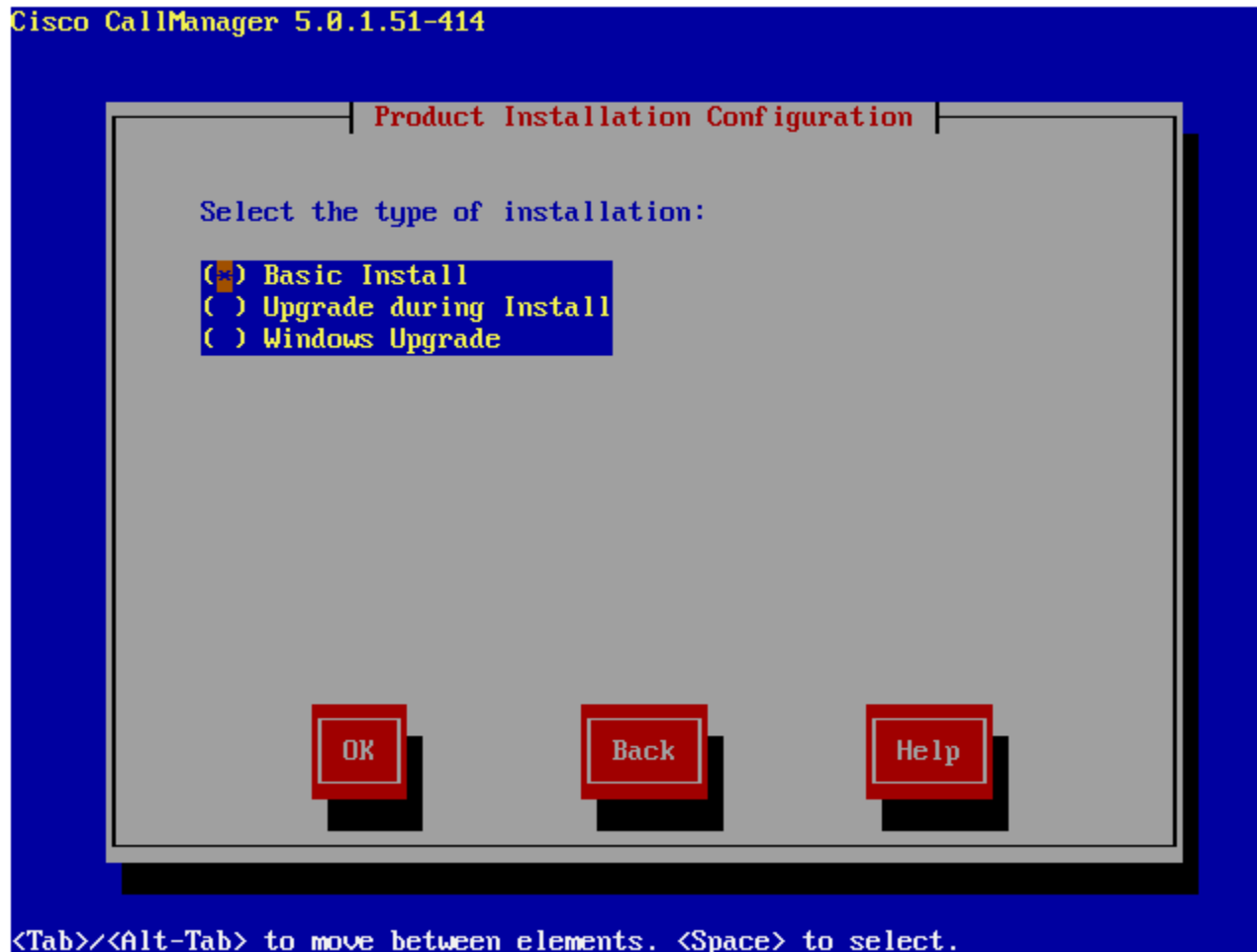
Skip

Cancel

<Tab>/<Alt-Tab> to move between elements. <Space> to select.



# Basic Installation Procedure



# Timezone Configuration

Cisco CallManager 5.0.1.51-414

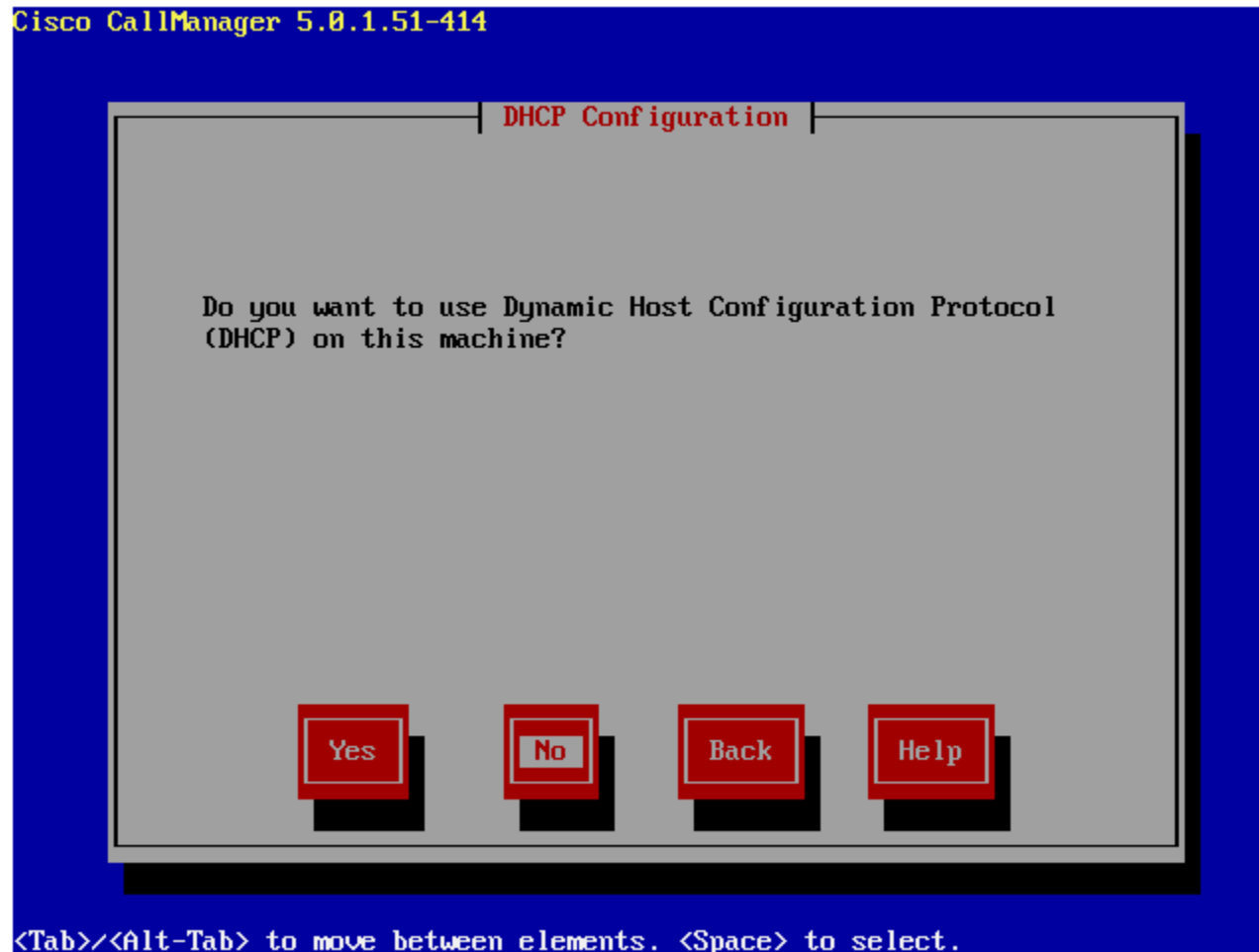
Timezone Configuraton

(GMT-12:00) Eniwetok, Kwajalein  
(GMT-11:00) Midway Island, Samoa #  
(GMT-10:00) Hawaii  
(GMT-09:00) Alaska  
**(GMT-08:00) Pacific Time (US & Canada)**  
(GMT-07:00) Arizona  
(GMT-07:00) Mountain Time (US & Canada)  
(GMT-06:00) Central Time (US & Canada)

OK                      Back                      Help

<Arrow Up/Down> to select, <Tab> to move to another field, <OK> to exit screen.

# Static Network Configuration



# Static Network Configuration

Cisco CallManager 5.0.1.51-414

Static Network Configuration

Host Name CM1

IP Address 192.168.1.10

IP Mask 255.255.255.0

GW Address 192.168.1.254

OK Back Help

<Tab>/<Alt-Tab> to move between elements. <Space> to select.

# DNS Client Configuration

Cisco CallManager 5.0.1.51-414

## DNS Client Configuration

Do you want to enable Domain Name System (DNS) Client  
on this machine?

Yes

No

Back

Help

<Tab>/<Alt-Tab> to move between elements. <Space> to select.

# Administrator Login Configuration

Cisco CallManager 5.0.1.51-414

Administrator Login Configuration

Administrator ID

Password

Confirm Password

<Tab>/<Alt-Tab> to move between elements. <Space> to select.

# Certificate Configuration

Cisco CallManager 5.0.1.51-414

Certificate Signing Request Configuration

Organization	Cisco Systems
Unit	WSFD
Location	San Jose
State	California
Country	Uganda Ukraine United Arab Emirates United States

OK Back Help

<Tab>/<Alt-Tab> to move between elements. <Space> to select.





# Database Access Security Configuration

Cisco CallManager 5.0.1.51-414

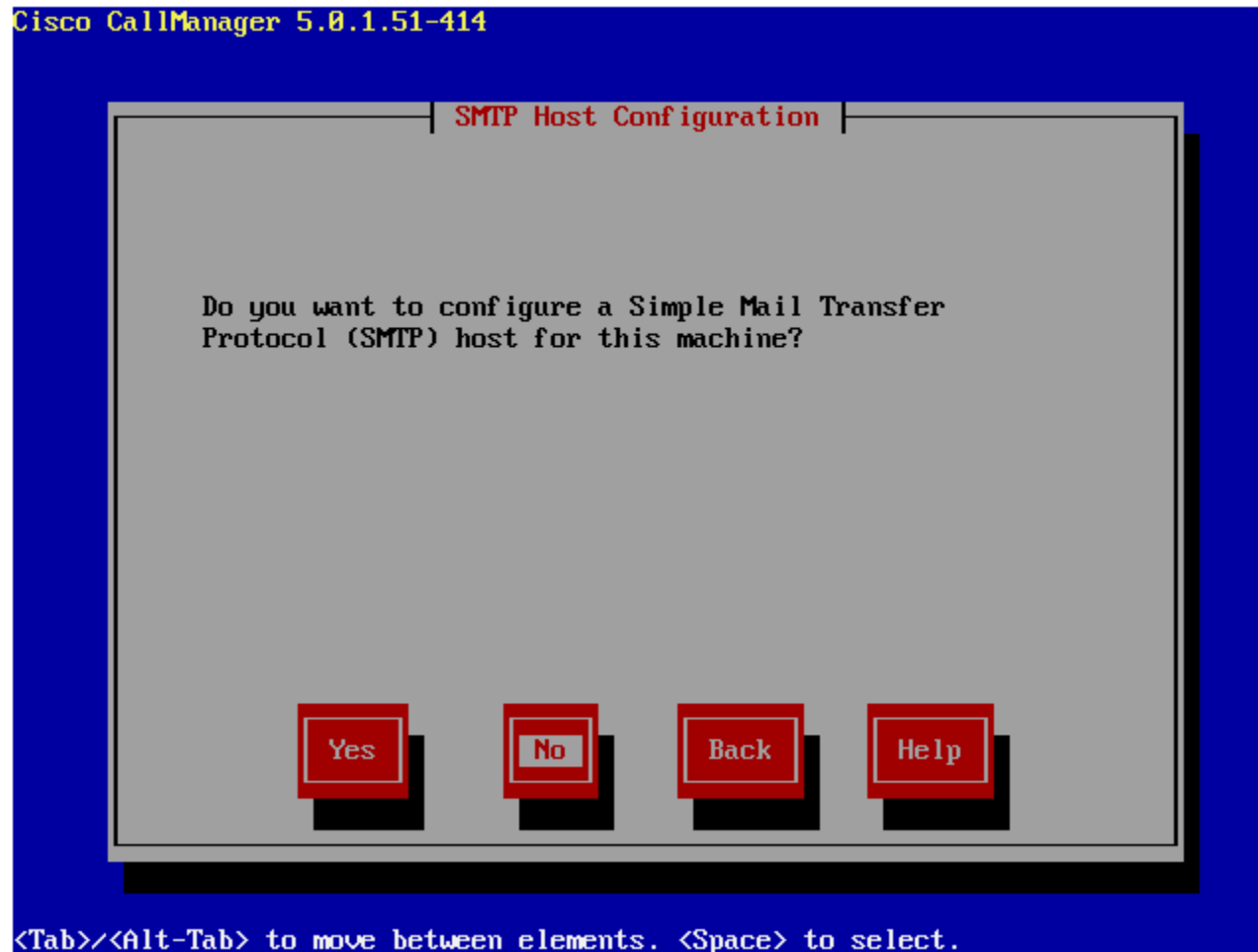
Database Access Security Configuration

Security Password

Confirm Password

<Tab>/<Alt-Tab> to move between elements. <Space> to select.

# SMTP Configuration



# CCMAdministrator Password Configuration

Cisco CallManager 5.0.1.51-414

**Application User Password Configuration**

The Application User Password is the password you use to log into the CCM administrative webpage as CCMAdministrator. This password also applies to all other predefined application users. The list of application users can be found by selecting Help on this page

Application User Password

Confirm Application User Password

<Tab>/<Alt-Tab> to move between elements. <Space> to select.

# Confirm Platform Configuration

Cisco CallManager 5.0.1.51-414

## Platform Configuration Confirmation

The Platform Configuration is complete.

Select OK to continue or Back to change the configuration.

Warning: Once you select OK, you will no longer be able to modify the Platform Configuration.

OK

Back

Cancel

<Tab>/<Alt-Tab> to move between elements. <Space> to select.

# Obtain License File

## License Calculation:

1. Obtain MAC address of publisher using platform administration.
2. Go to CCO and <http://www.cisco.com/go/license>. T. E. 1
3. Use software serial number / PAK and MAC address to obtain license file.

**Slide 21**

---

**T. E.1**      **Need clarification by Cisco.**  
Tolga Erdogan, 3/14/2006

# Obtain License File: Publisher MAC Address

To access Platform Administration go to:  
<https://<IP>/iptplatform>

**Logon**

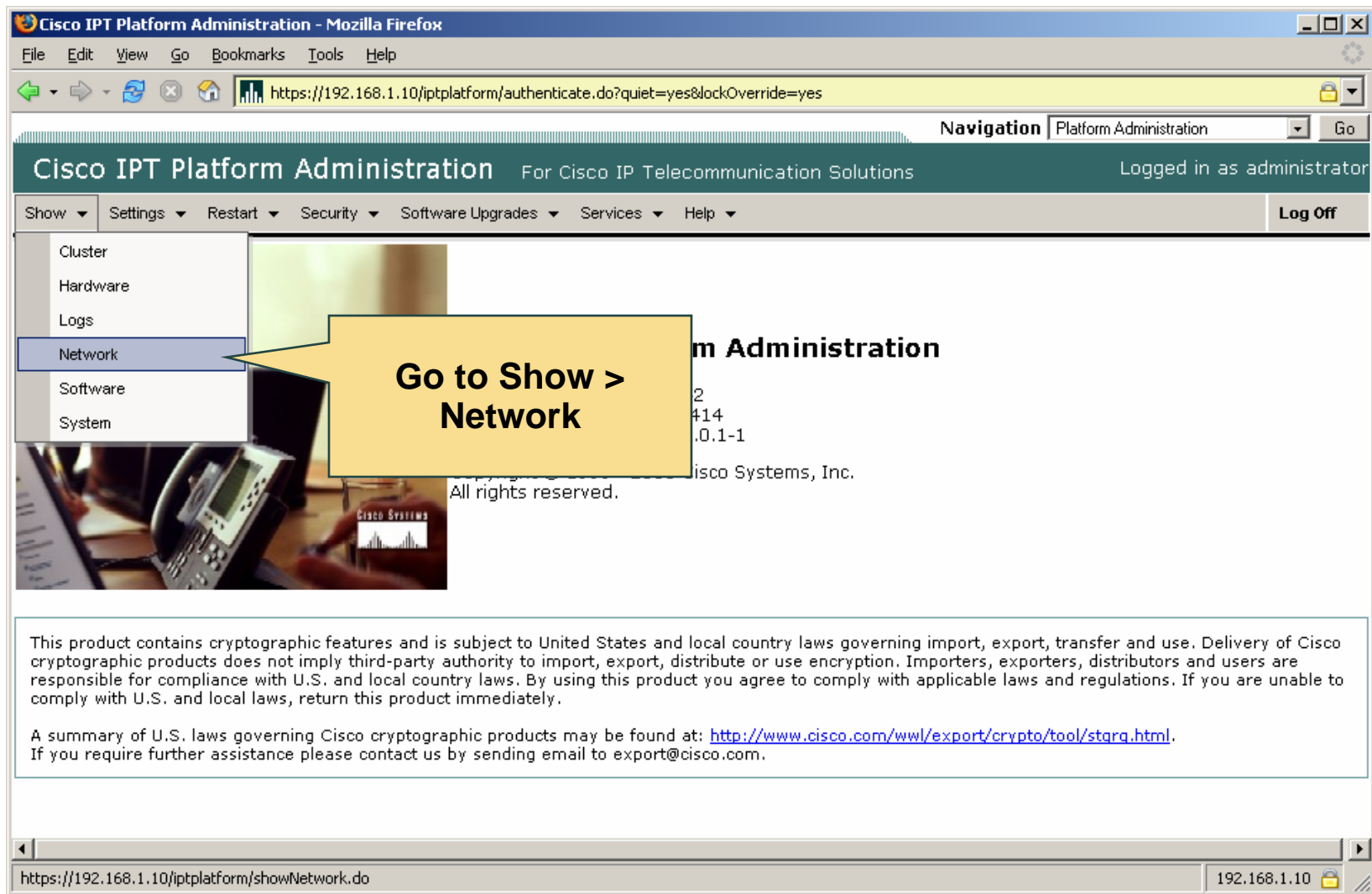
Username administrator  
Password

Submit Reset

**Login:  
administrator**

Done 192.168.1.10

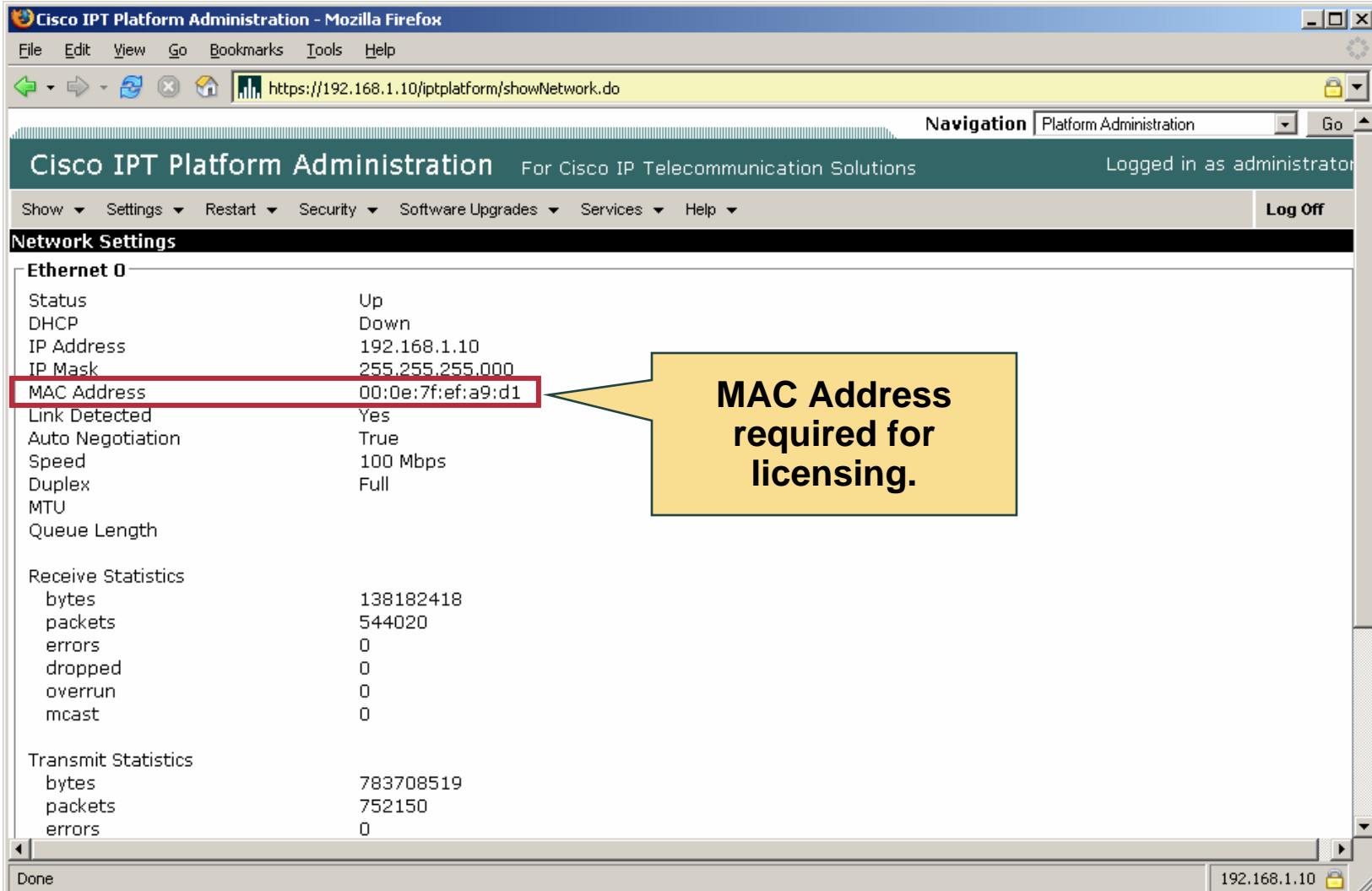
# Obtain License File: Publisher MAC Address



The screenshot shows a Mozilla Firefox browser window displaying the Cisco IPT Platform Administration interface. The address bar shows the URL `https://192.168.1.10/iptplatform/authenticate.do?quiet=yes&lockOverride=yes`. The page title is "Cisco IPT Platform Administration" and it indicates the user is logged in as "administrator". A navigation menu is visible with options: Show, Settings, Restart, Security, Software Upgrades, Services, and Help. The "Show" menu is expanded, showing sub-options: Cluster, Hardware, Logs, Network (highlighted), Software, and System. A yellow callout box with a speech bubble points to the "Network" option, containing the text "Go to Show > Network". The main content area shows a partial view of the "System Administration" page with some text and a Cisco logo. The status bar at the bottom of the browser shows the current URL as `https://192.168.1.10/iptplatform/showNetwork.do` and the IP address `192.168.1.10`.



# Obtain License File: Publisher MAC Address



The screenshot shows the Cisco IPT Platform Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10/iptplatform/showNetwork.do`. The page title is "Cisco IPT Platform Administration" and it indicates the user is logged in as "administrator".

The "Network Settings" section is expanded to show details for "Ethernet 0". The following table represents the configuration data shown in the interface:

Parameter	Value
Status	Up
DHCP	Down
IP Address	192.168.1.10
IP Mask	255.255.255.000
MAC Address	00:0e:7f:ef:a9:d1
Link Detected	Yes
Auto Negotiation	True
Speed	100 Mbps
Duplex	Full
MTU	
Queue Length	
Receive Statistics	
bytes	138182418
packets	544020
errors	0
dropped	0
overrun	0
mcast	0
Transmit Statistics	
bytes	783708519
packets	752150
errors	0

A yellow callout bubble with a black border points to the "MAC Address" field, containing the text: "MAC Address required for licensing."

# Install License File

## Installing License Files:

1. Access CCMAdmin Webpage
2. Upload license file to Cisco Unified CallManager.
3. Review license status.

# Install License File: Access CCMAdmin Webpage

**To access the administrative GUI go to:  
https://<IP>/ccmadmin**

**Login:  
CCMAdministrator**

Done 192.168.1.10

# Install License File: Upload License File

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccadmin/serverFindList.do`. The page title is "Cisco CallManager Administration" and it is logged in as "CCMAdministrator". The navigation menu includes "Server", "Cisco CallManager", "Cisco CallManager Group", "Phone NTP Reference", "Date/Time Group", "Presence Group", "Region", "Device Pool", "DHCP", "LDAP", "Location", "SRST", "MLPP Domain", "Enterprise Parameters", "Service Parameters", "Security Profile", "Application Server", and "Licensing". The "Licensing" menu item is expanded, showing "License Unit Report", "License Unit Calculator", and "License File Upload". A yellow callout box with a circled "1" points to the "License File Upload" option, containing the text: "Go to Server > Licensing > License File Upload". The browser address bar at the bottom shows `https://192.168.1.10:8443/ccadmin/licensingFileUploadEdit.do`.

# Install License File: Upload License File

The screenshot displays the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10:8443/ccmadmin/licensingFileUploadEdit.do`. The page title is "License File Upload".

**Step 2:** A yellow callout box with the number "2" points to the "Upload License File" button in the "License File Information" section. The text inside the callout reads: "Click on Upload License File."

**Step 3:** A second yellow callout box with the number "3" points to the "Upload" button in the "Upload File" dialog box. The text inside the callout reads: "Specify License File location and click upload." The "File:" field in the dialog box contains the path `c:\ccm50license\CCM20051220173627639.lic`.

# Install License File: Verify License Status

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

Server Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Server

- Cisco CallManager
- Cisco CallManager Group
- Phone NTP Reference
- Date/Time Group
- Presence Group
- Region
- Device Pool
- DHCP
- LDAP
- Location
- SRST
- MLPP Domain
- Enterprise Parameters
- Service Parameters
- Security Profile
- Application Server
- Licensing

Cisco CallManager Administration

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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All rights reserved.

License Unit Report

License Unit Calculator

License File Upload

Go to Server > Licensing > License Unit Report.

https://192.168.1.10:8443/ccmadmin/licensingreportEdit.do 192.168.1.10:8443

# Install License File: Verify License Status

The screenshot shows the Cisco CallManager Administration interface. The page title is "License Unit Report". The status is "Ready". The page is divided into two main sections: "Phone License Feature" and "CCM Node License Feature". Each section contains a table with columns for "License Server", "Units Authorized", "Units Used", and "Units Remaining".

**Phone License Feature**

License Server	Units Authorized	Units Used	Units Remaining
192.168.1.10	100	11	89
<b>Total Units for Feature</b>	<b>100</b>	<b>11</b>	<b>89</b>

**CCM Node License Feature**

License Server	Units Authorized	Units Used	Units Remaining
192.168.1.10	3	1	2
<b>Total Units for Feature</b>	<b>3</b>	<b>1</b>	<b>2</b>

Callouts from the image:

- Device Units authorized. (points to 100)
- Device Units used. (points to 11)
- Device Units remaining. (points to 89)
- Node Units authorized. (points to 3)
- Node Units used. (points to 1)
- Node Units remaining. (points to 2)

# Summary

- **Cisco Unified CallManager 5.0 has strict hardware requirements which need to be met.**
- **Additional information, e.g. IP address, DNS and NTP should be available prior to installation.**
- **The actual installation is done via a wizard.**
- **A license file needs to be obtained using the publisher MAC address.**
- **Install a license file prior to actual configuration.**



# CISCO SYSTEMS





**Deploying a Cisco Unified CallManager 5.0 Cluster**

**Upgrading to Cisco Unified CallManager 5.0**

# Objectives

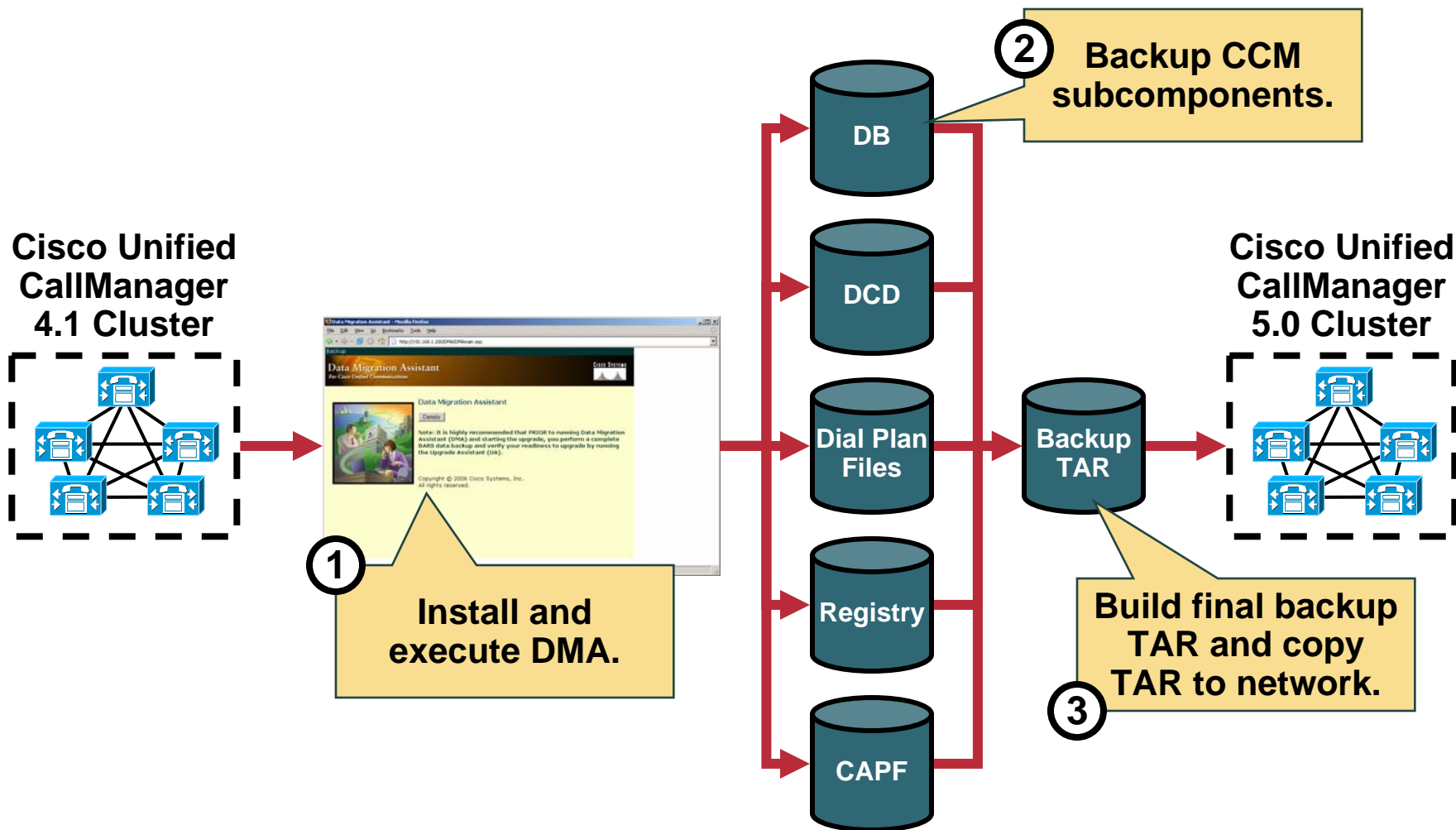
- **Data Migration Assistant Overview**
- **Installing the Data Migration Assistant**
- **Backing up Cisco Unified CallManager 4.1 deployments**
- **Performing an Upgrade Installation**

# Data Migration Assistant Overview

**The Data Migration Assistant (DMA) is used to back-up CCM 4.1 for migrations:**

- **Installed on a CCM 4.1 publisher.**
- **Similar to BARS**
  - **Same look and feel.**
  - **Same procedure.**
- **Back-up used during CCM 5.0 upgrade installation.**
- **DMA can be obtained from CCO:**  
**<http://www.cisco.com/kobayashi/sw-center/sw-voice.shtml>**

# Data Migration Assistant Overview



# Data Migration Assistant Overview: Caveats

## **DMA won't back up:**

- **Custom TFTP phone loads and Music On Hold (MOH) files**
- **CCMUser password and PIN number**
- **Customer installed Third Party Products**
- **Customer installed Phone Background Images**

## **User password is reset to ciscocisco**

- **Default can be configured during CCM5.0 Install**

## **User EM / IPMA pin is reset to 12345**

- **Default can be configured during CCM5.0 Install**

# DMA Installation

## **DMA is installed on a publisher:**

- **Install DMA on a CCM 4.1 publisher.**
- **No installation options available.**
- **Server will require post-installation reboot:**
  - **Never install DMA during production hours.**
- **Access DMA via Web GUI:**
  - **http://<ip of CCM>/DMA**
  - **Same login as in CCM Administration.**

# DMA Back-Up Procedure

## **DMA back-up steps:**

- 1. Backup the existing 4.x deployment using BARS.**
- 2. Verify integrity using the Upgrade Assistant.**
- 3. Access DMA GUI.**
- 4. Specify back-up location.**
- 5. Perform backup.**
- 6. Copy TAR to a network directory (optional).**



# DMA Back-Up Procedure: Access DMA

Backup

## Data Migration Assistant

For Cisco Unified Communications

CISCO SYSTEMS

**Access DMA URL:  
http://<ip>/DMA**

Data Migration Assistant

Details

**Note: It is highly recommended that PRIOR to running Data Migration Assistant (DMA) and starting the upgrade, you perform a complete BARS data backup and verify your readiness to upgrade by running the Upgrade Assistant (UA).**

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All rights reserved.

Done

# DMA Back-Up Procedure: Specify Storage Location

The screenshot shows the 'Data Migration Assistant - Backup Storage Location' web interface in a Mozilla Firefox browser. The page has a navigation menu with 'Home', 'Storage Location', and 'Backup Now'. A 'Backup Storage Location' section contains an 'Update' button. Below this are 'Destination Options' with radio buttons for 'Network Directory', 'Local Directory', and 'Tap'. The 'Local Directory' option is selected. Under 'Local Directory', there is a 'Path Name' field containing 'c:\backup' and a 'Browse' button. A dropdown menu below shows 'Not Selected'. Two yellow callout boxes provide instructions: callout 1 points to the 'Storage Location' menu item with the text 'Go to Backup > Storage Location.'; callout 2 points to the 'Local Directory' radio button with the text 'Select local directory as storage location.'; callout 3 points to the 'Path Name' field with the text 'Specify directory and path.'.

Two dialog boxes are overlaid on the page:

- Dialog 4 (top right): A warning dialog with a yellow triangle icon. Text: 'Please ensure that you transfer the contents from the LOCAL path to an external device before upgrading. The files will not be readable from the local directory during the upgrade installation.' An 'OK' button is at the bottom right.
- Dialog 5 (bottom right): A success dialog with a yellow triangle icon. Text: 'Storage location successfully updated.' An 'OK' button is at the bottom right.

Numbered callouts 1 through 5 are placed around the interface to guide the user through the steps.

# DMA Back-Up Procedure: Perform Backup Procedure

**1** Go to Backup > Backup Now.

**2** Click on Start Backup Now

Verify the current backup status.

**Data Migration Assistant - Migrate Data Source Server - Mozilla Firefox**

File Edit View Go Bookmarks Tools Help

Backup

- Home
- Storage Location
- Backup Now

**Migrate Data Source Se**

Start Backup Now (View Latest Status)

Status: Ready

**Migration Assistant - Current Status**

ted at 02:46:10 PM on 02/20/2006.

[M] DMA version is 5.0.1000.

[02:46:10 PM] Local machine name is Cm1.

[02:46:10 PM] Local IP Address is 10.128.31.101.

[02:46:10 PM] Service account is Dmaback\_Admin.

[02:46:35 PM] Dmaback\_Admin has local administrative rights.

[02:46:35 PM] Backup running with low priority.

[02:46:35 PM] Backup Timestamp is 02-20-06#14-46.

[02:46:35 PM] Found 59534MB available on the boot partition for staging the backup.

[02:46:46 PM] Beginning Cisco CallManager backup from 10.128.31.101.

[02:46:46 PM] Backing up Cisco CallManager databases.

[02:46:46 PM] This can take a long time, depending on the size of the databases. Be patient.

[02:47:16 PM] Exportable format of Cisco CallManager databases backed up successfully.

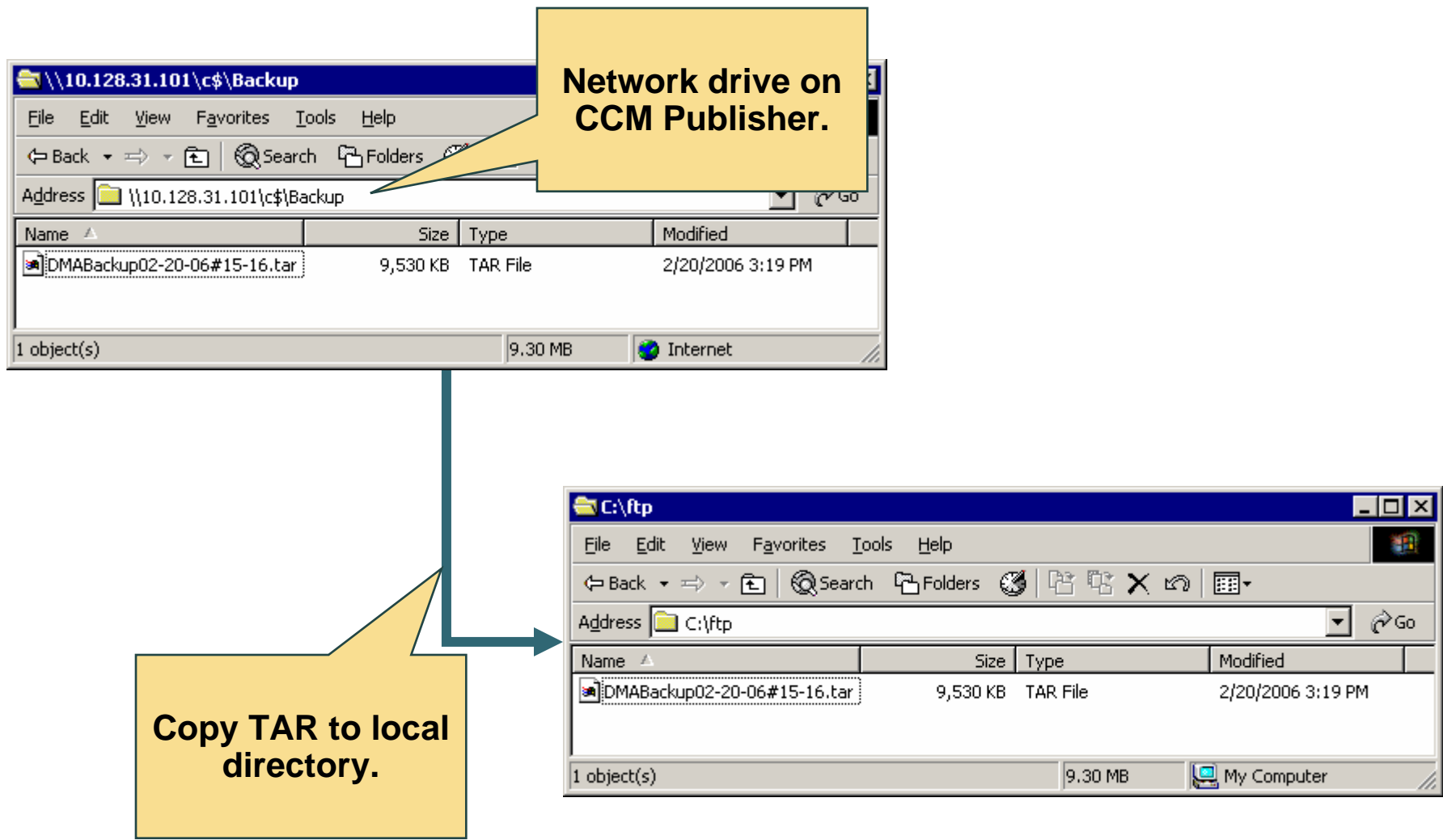
[02:47:16 PM] Backing up Cisco CallManager directories.

[02:47:16 PM] This can take a long time, depending on the size of the directories. Be patient.

http://192.168.1.200/DMA/DMAbackup.asp

Done

# DMA Back-Up Procedure: Copy TAR

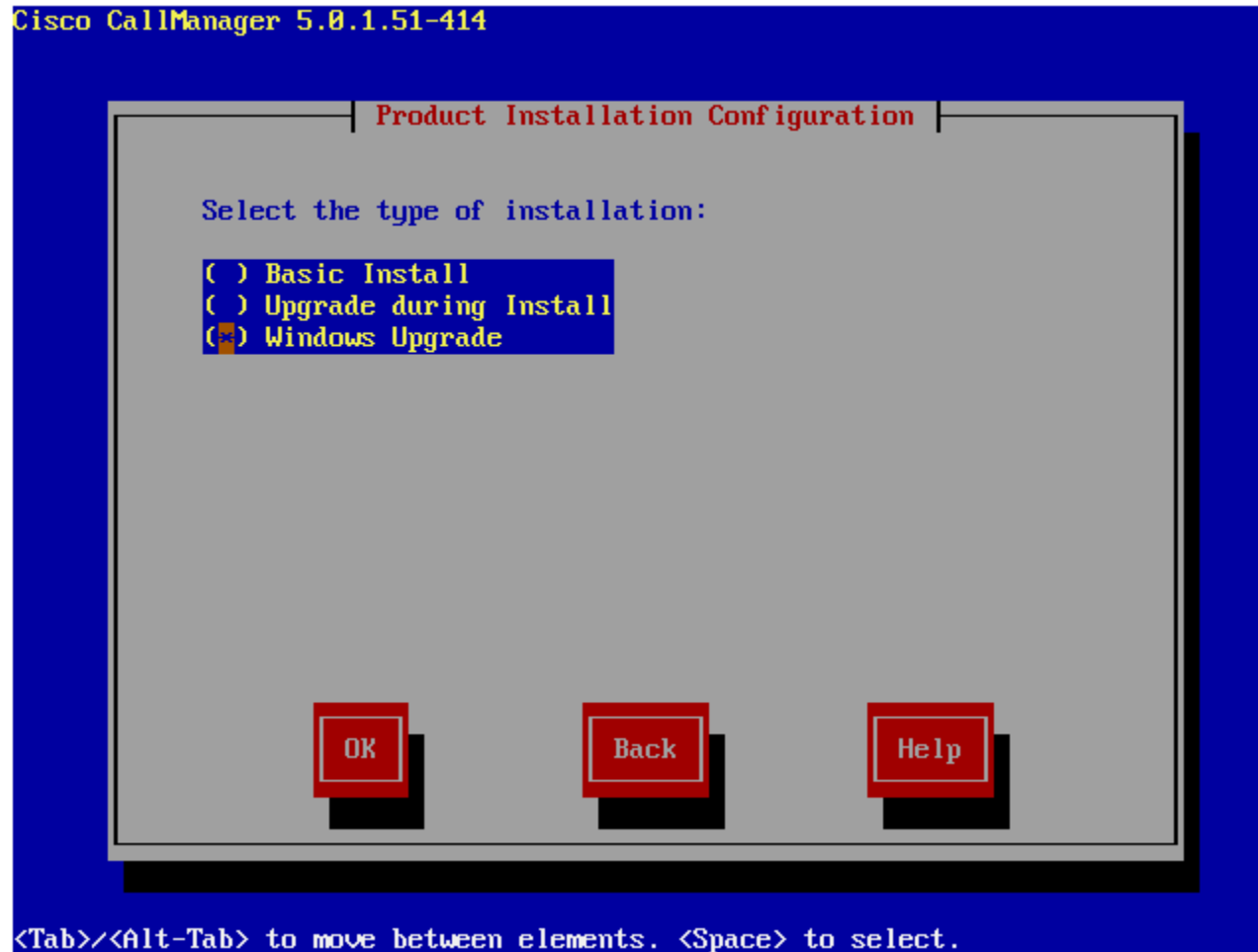


# Upgrade Procedure

**Perform installation procedure using the upgrade option:**

- 1. Run installer with upgrade option.**
- 2. When prompted, specify FTP location of TAR file.**

# Upgrade Procedure



# Summary

- **DMA backs up existing CCM 4.1 deployments for migration and is similar to BARS.**
- **DMA is installed on the publisher server and requires a reboot.**
- **Backup can be stored on a network, local or tape drive.**
- **The DMA TAR is used during an upgrade installation to pre-provision the database.**

# CISCO SYSTEMS







## **Administering Cisco Unified CallManager 5.0**

# **Performing General Administration**

# Objectives

- **General Administration Overview**
- **Accessing Cisco Unified CallManager 5.0**
- **Cisco Unified CallManager 5.0 System Menu**
- **Cisco Unified CallManager 5.0 Call Routing Menu**
- **Cisco Unified CallManager 5.0 Media Resources Menu**
- **Cisco Unified CallManager 5.0 Voice Mail Menu**
- **Cisco Unified CallManager 5.0 Device Menu**
- **Cisco Unified CallManager 5.0 Application Menu**
- **Cisco Unified CallManager 5.0 User Management Menu**
- **Multi Level Admin**

# General Administration Overview

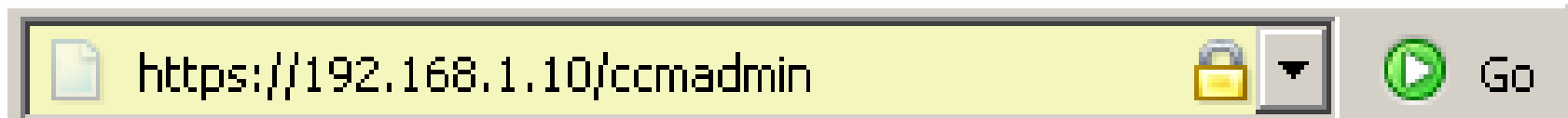
## Administration of Cisco Unified CallManager 5.0

- Complete “Out of the box” system without direct access to the operating system.
- Web-based Cisco Unified CallManager administration using HTTPS to secure the session.
- SSH or web-based platform administration.

# Accessing Cisco Unified CallManager 5.0

**To enter the Cisco Unified CallManager administration page go to URL:**

- **Directly access the log-in page by going to `https://<CallManager IP>/ccmadmin`**
- **Getting a link that leads to the log-in page by going to `http://<CallManager IP>`**



# Accessing Cisco Unified CallManager 5.0

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

CISCO SYSTEMS

Cisco CallManager Administration For Cisco IP Telecommunication Solutions

Logon

Username: CCMAdministrator

Password: [masked]

Submit Reset

Use CCMAdministrator as username

Enter the password configured during the installation process

Note: Username and password are case sensitive!

Done 192.168.1.10

# Cisco Unified CallManager 5.0 System Menu

**The system menu is used to configure:**

- **Overall server parameters.**
- **General device settings and groups.**
- **Locations and regions.**
- **Cluster wide settings and service parameters.**

# Cisco Unified CallManager 5.0 System Menu: Server Parameters

The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser window. The browser's address bar displays the URL `https://192.168.1.10/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The navigation menu at the top includes: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "System" menu is expanded, showing a list of server-related parameters:

- Server
- Cisco CallManager
- Cisco CallManager Group
- Phone NTP Reference
- Date/Time Group
- Presence Group
- Region
- Device Pool
- DHCP
- LDAP
- Location
- SRST
- MLPP Domain
- Enterprise Parameters
- Service Parameters
- Security Profile
- Application Server
- Licensing

The main content area displays the "Cisco CallManager Administration" header, system version (5.0.1.51-414), administration version (1.1.0.0-1), and copyright information (© 1999 - 2005 Cisco Systems, Inc.). A legal disclaimer is also visible, stating that the product is subject to U.S. and local country laws governing import, export, transfer, and use.

# Cisco Unified CallManager 5.0 System Menu: Device Settings

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

- Server
  - Cisco CallManager
  - Cisco CallManager Group
  - Phone NTP Reference
  - Date/Time Group
  - Presence Group
  - Region
  - Device Pool
  - DHCP
  - LDAP
  - Location
  - SRST
  - MLPP Domain
  - Enterprise Parameters
  - Service Parameters
  - Security Profile
  - Application Server
  - Licensing

**Cisco CallManager Administration**

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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...e please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).



# Cisco Unified CallManager 5.0 System Menu: Location Based Settings

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

- Server
- Cisco CallManager
- Cisco CallManager Group
- Phone NTP Reference
- Date/Time Group
- Presence Group
- Region
- Device Pool
- DHCP
- LDAP
- Location
- SRST
- MLPP Domain
- Enterprise Parameters
- Service Parameters
- Security Profile
- Application Server
- Licensing

**Cisco CallManager Administration**

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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...e please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

# Cisco Unified CallManager 5.0 System Menu: Cluster Settings

The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser window. The browser's address bar displays the URL `https://192.168.1.10/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The navigation menu at the top includes: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. A "Log Off" button is also present.

The "System" menu is expanded, showing a list of settings categories. The following items are highlighted with a red grid pattern:

- DHCP
- LDAP
- Enterprise Parameters
- Service Parameters
- Application Server
- Licensing

The main content area of the page displays the following information:

### Cisco CallManager Administration

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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Below this information, there is a section containing legal notices regarding cryptographic features and import/export regulations, including a link to <http://www.cisco.com/wwl/export/crypto/tool/stqrg.html>.

# Cisco Unified CallManager 5.0

## Call Routing Menu

**The call routing menu includes these functionalities:**

- **Dial plan and configuration.**
- **Call routing permission and restriction configuration.**
- **Number handling feature configuration.**
- **Tools to analyze and install dial plans.**

# Cisco Unified CallManager 5.0 Call Routing Menu: Dial Plan

The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser window. The browser's address bar displays the URL `https://192.168.1.10/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The navigation menu at the top includes: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "Call Routing" menu is expanded, showing a list of options: A&R Group, Dial Rules, Route Filter, Route/Hunt, SIP Route Pattern, Class of Control, Client Matter Codes, Forced Authorization Codes, Translation Pattern, Call Park, Call Pickup Group, Directory Number, Meet-Me Number/Pattern, Route Plan Report, and Dial Plan Installer. The "Dial Plan" section is highlighted with a red grid pattern.

The main content area displays "Cisco CallManager Administration" with system and administration versions (5.0.1.51-414 and 1.1.0.0-1), copyright information (© 1999 - 2005 Cisco Systems, Inc.), and a disclaimer regarding import, export, and use of cryptographic products. A link to <http://www.cisco.com/wwl/export/crypto/tool/stqrg.html> is provided for cryptographic products.

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

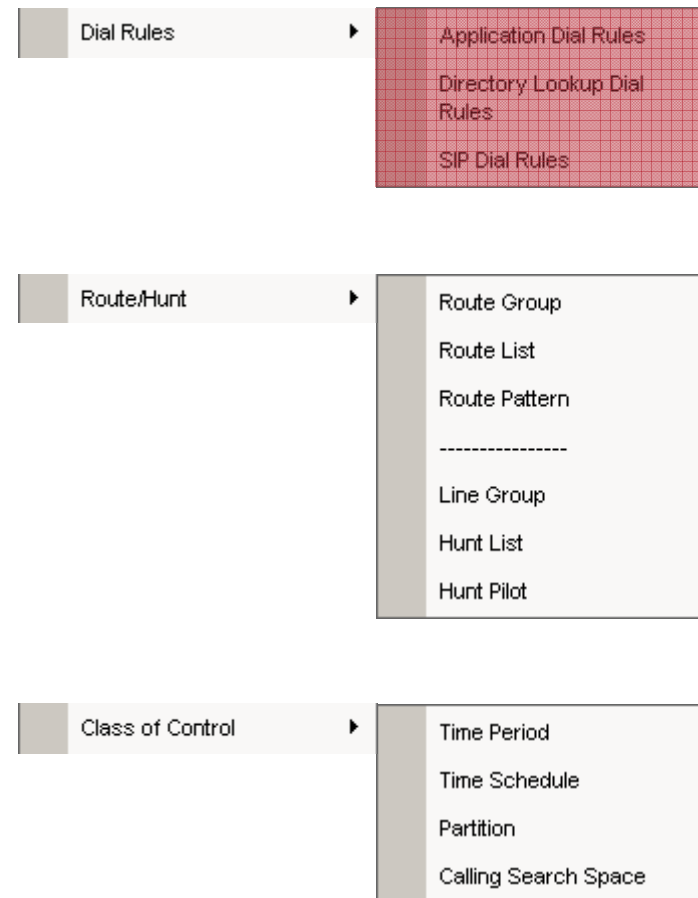




# Cisco Unified CallManager 5.0 Call Routing Menu

**Dial rules are used to modify numbers which are dialed using:**

- External applications like IPMA.
- Directory lookup.
- A special SIP device.

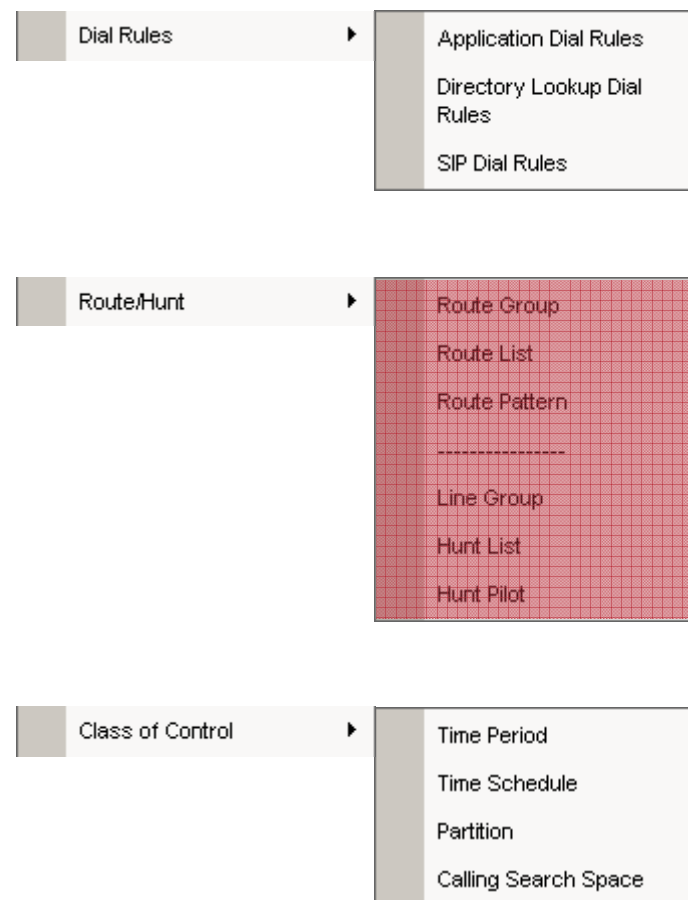


# Cisco Unified CallManager 5.0

## Call Routing Menu

**The route/hunt sub menu includes two parts:**

- **Route group/list/patterns to route calls outside the cluster.**
- **Line group and hunt list/pilot to distribute calls within the cluster.**

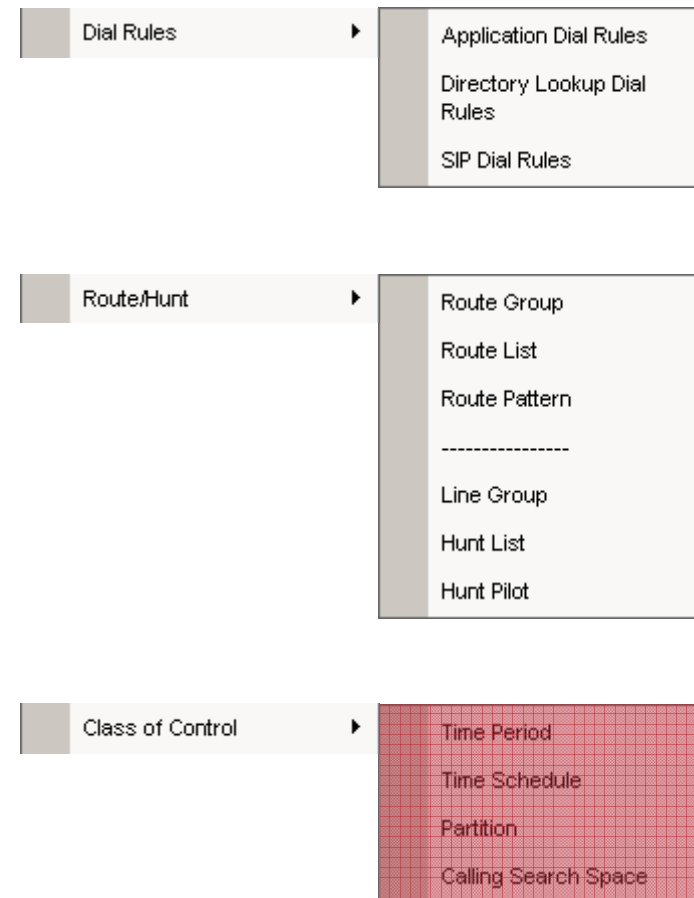




# Cisco Unified CallManager 5.0 Call Routing Menu

**The class of control sub menu is used to define:**

- **Time period/schedule for time based call routing.**
- **Partitions and calling search spaces to create Class of Restriction (COR) based dial plans.**



# Cisco Unified CallManager 5.0

## Media Resources Menu

**The media resources menu is used to manage media applications and devices:**

- **Annunciators that are used to play back .wav files on phones and gateways.**
- **Software and hardware conference bridges.**
- **Media termination point (MTP) to terminate media e.g. for protocol translations between SIP and SCCP.**
- **MOH servers and sources.**
- **External transcoders to relieve MTP.**
- **Media resource groups and group lists to make media resources available to devices.**
- **MOH file management to upload/configure individual MOH.**

# Cisco Unified CallManager 5.0 Media Resources Menu: Annunciator

The screenshot shows the Cisco CallManager Administration web interface. The browser window title is "Cisco CallManager Console - Mozilla Firefox". The address bar shows the URL "https://192.168.1.10/ccmadmin/showHome.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu is expanded to "Media Resources", and the "Annunciator" option is selected and highlighted in red. The main content area displays the following information:

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- Administration version: 1.1.0.0-1
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The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

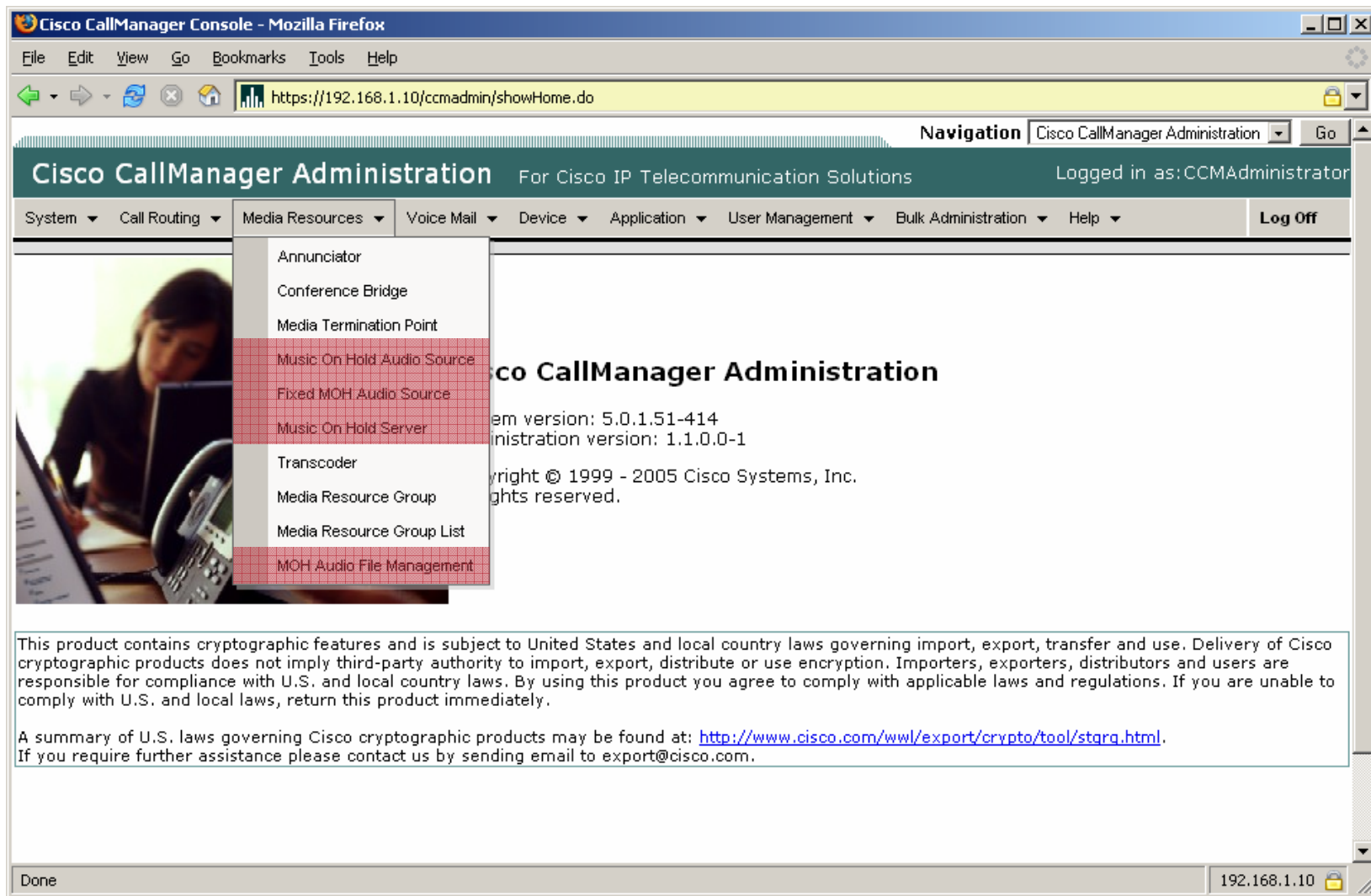
# Cisco Unified CallManager 5.0 Media Resources Menu: Conference Bridge

The screenshot shows the Cisco CallManager Administration web interface. The browser window title is "Cisco CallManager Console - Mozilla Firefox". The address bar shows the URL "https://192.168.1.10/ccmadmin/showHome.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu is open, showing the following options: System, Call Routing, Media Resources (selected), Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "Media Resources" dropdown menu is open, showing the following options: Annunciator, Conference Bridge (highlighted), Media Termination Point, Music On Hold Audio Source, Fixed MOH Audio Source, Music On Hold Server, Transcoder, Media Resource Group, Media Resource Group List, and MOH Audio File Management. The main content area displays "Cisco CallManager Administration" and the following information: "System version: 5.0.1.51-414", "Administration version: 1.1.0.0-1", and "Copyright © 1999 - 2005 Cisco Systems, Inc. All rights reserved." Below this is a legal disclaimer: "This product contains cryptographic features and is subject to United States and local country laws governing import, export, transfer and use. Delivery of Cisco cryptographic products does not imply third-party authority to import, export, distribute or use encryption. Importers, exporters, distributors and users are responsible for compliance with U.S. and local country laws. By using this product you agree to comply with applicable laws and regulations. If you are unable to comply with U.S. and local laws, return this product immediately." A link is provided: "A summary of U.S. laws governing Cisco cryptographic products may be found at: <http://www.cisco.com/wwl/export/crypto/tool/stqrg.html>. If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com)." The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 Media Resources Menu: Media Termination Point

The screenshot shows the Cisco CallManager Administration web interface. The browser window title is "Cisco CallManager Console - Mozilla Firefox". The address bar shows the URL "https://192.168.1.10/ccmadmin/showHome.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu is open, with "Media Resources" selected, and a dropdown menu is visible. The dropdown menu items are: Annunciator, Conference Bridge, Media Termination Point (highlighted), Music On Hold Audio Source, Fixed MOH Audio Source, Music On Hold Server, Transcoder, Media Resource Group, Media Resource Group List, and MOH Audio File Management. The main content area displays "Cisco CallManager Administration" and system information: "System version: 5.0.1.51-414" and "Administration version: 1.1.0.0-1". Below this is the copyright notice: "Copyright © 1999 - 2005 Cisco Systems, Inc. All rights reserved." A legal disclaimer is also present, stating that the product contains cryptographic features and is subject to U.S. and local country laws. A link is provided for more information: <http://www.cisco.com/wwl/export/crypto/tool/stqrg.html>. The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 Media Resources Menu: Music On Hold

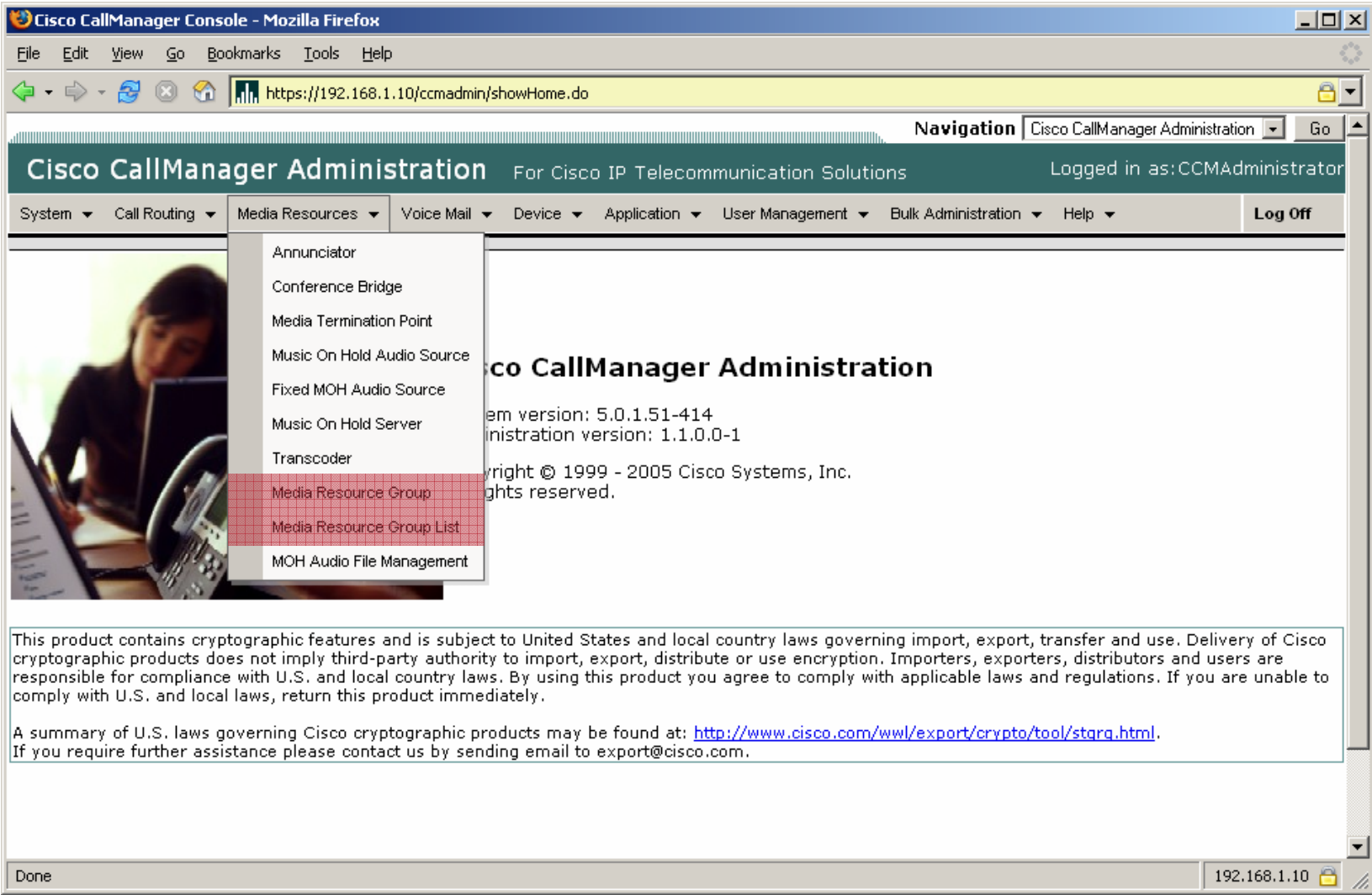


The screenshot shows the Cisco CallManager Administration web interface. The browser window title is "Cisco CallManager Console - Mozilla Firefox". The address bar shows the URL "https://192.168.1.10/ccmadmin/showHome.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu is expanded to "Media Resources", which has opened a dropdown menu with the following items: Annunciator, Conference Bridge, Media Termination Point, Music On Hold Audio Source (highlighted with a red grid), Fixed MOH Audio Source, Music On Hold Server, Transcoder, Media Resource Group, Media Resource Group List, and MOH Audio File Management (highlighted with a red grid). The main content area displays "Cisco CallManager Administration" and system information: "System version: 5.0.1.51-414", "Administration version: 1.1.0.0-1", and "Copyright © 1999 - 2005 Cisco Systems, Inc. All rights reserved." A legal disclaimer is present at the bottom of the page, and the status bar at the bottom of the browser shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 Media Resources Menu: Transcoder

The screenshot shows the Cisco CallManager Administration web interface. The browser window title is "Cisco CallManager Console - Mozilla Firefox". The address bar shows the URL "https://192.168.1.10/ccmadmin/showHome.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Media Resources" menu is open, showing options: "Annunciator", "Conference Bridge", "Media Termination Point", "Music On Hold Audio Source", "Fixed MOH Audio Source", "Music On Hold Server", "Transcoder" (highlighted), "Media Resource Group", "Media Resource Group List", and "MOH Audio File Management". The main content area displays "Cisco CallManager Administration" and system information: "System version: 5.0.1.51-414" and "Administration version: 1.1.0.0-1". A legal disclaimer is present at the bottom of the page, stating: "This product contains cryptographic features and is subject to United States and local country laws governing import, export, transfer and use. Delivery of Cisco cryptographic products does not imply third-party authority to import, export, distribute or use encryption. Importers, exporters, distributors and users are responsible for compliance with U.S. and local country laws. By using this product you agree to comply with applicable laws and regulations. If you are unable to comply with U.S. and local laws, return this product immediately. A summary of U.S. laws governing Cisco cryptographic products may be found at: <http://www.cisco.com/wwl/export/crypto/tool/stqrg.html>. If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com)." The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 Media Resources Menu: Media Resources





# Cisco Unified CallManager 5.0

## Voice Mail Menu

**The voice mail menu is used to manage Voicemail integration including:**

- **Unity voice mail ports.**
- **Message Waiting Indication (MWI).**
- **Voice mail pilot number.**
- **Voice mail profiles.**

# Cisco Unified CallManager 5.0 Voice Mail Menu: Voice Port Configuration



Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

- Cisco Voice Mail Port
- Cisco Voice Mail Port Wizard
- Message Waiting
- Voice Mail Pilot
- Voice Mail Profile

CallManager Administration

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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Done 192.168.1.10

# Cisco Unified CallManager 5.0 Voice Mail Menu: Message Waiting Indication



Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

- Cisco Voice Mail Port
- Cisco Voice Mail Port Wizard
- Message Waiting
- Voice Mail Pilot
- Voice Mail Profile

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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Done 192.168.1.10

# Cisco Unified CallManager 5.0

## Voice Mail Menu: Voice Mail Access



Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

- Cisco Voice Mail Port
- Cisco Voice Mail Port Wizard
- Message Waiting
- Voice Mail Pilot**
- Voice Mail Profile

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

Done 192.168.1.10

# Cisco Unified CallManager 5.0 Device Menu

**The device menu contains all necessary parts to configure devices:**

- **Logical devices:**
  - IP trunks to other systems.
  - CTI route points for CTI applications.
- **Physical devices**
  - Phones
  - Voice gateways
  - Gatekeepers
- **Device defaults and templates.**

# Cisco Unified CallManager 5.0

## Device Menu: Physical and Logical Devices

The screenshot displays the Cisco CallManager Administration web interface in a Mozilla Firefox browser window. The browser's address bar shows the URL `https://192.168.1.10/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The navigation menu includes: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "Device" menu is expanded, showing a list of device types: CTI Route Point, Gatekeeper, Gateway, Phone, and Trunk. A sub-menu for "Device Settings" is also visible, listing options such as Device Defaults, Firmware Load Information, Default Device Profile, Device Profile, Phone Button Template, Softkey Template, Phone Services, SIP Profile, and Common Phone Profile.

The main content area features a photograph of a woman working at a desk with a laptop and a Cisco IP phone. Below the photo, there is a copyright notice: "Copyright © 1999 - 2005 Cisco Systems, Inc. All rights reserved." and a disclaimer regarding cryptographic features and U.S. laws. A link to a summary of U.S. laws is provided: <http://www.cisco.com/pto/tool/stqrg.html>.

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 Device Menu: Profiles, Defaults, and Templates



The screenshot shows the Cisco CallManager Administration console in a Mozilla Firefox browser window. The browser's address bar displays the URL `https://192.168.1.10/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The navigation menu includes: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "Device" menu is expanded, showing the following options:

- CTI Route Point
- Gatekeeper
- Gateway
- Phone
- Trunk
- Device Settings (with a right-pointing arrow)

The "Device Settings" sub-menu is highlighted in red and contains the following items:

- Device Defaults
- Firmware Load Information
- Default Device Profile
- Device Profile
- Phone Button Template
- Softkey Template
- Phone Services
- SIP Profile
- Common Phone Profile

The main content area features a photograph of a woman working at a computer with a Cisco IP phone in the foreground. Below the photo, there is a copyright notice: "Copyright © 1999 - 2005 Cisco Systems, Inc. All rights reserved." and a disclaimer regarding cryptographic features and export laws. A link is provided for more information: <http://www.cisco.com/warp/cto/tool/stqrg.html>.

The status bar at the bottom of the browser window shows "Done" on the left and "192.168.1.10" on the right.

# Cisco Unified CallManager 5.0 Application Menu

**The application menu is used to manage Cisco Unified CallManager 5.0 applications:**

- **Configure Cisco IP Manager Assistant (IPMA).**
- **Configure Cisco Unified CallManager Attendant Console.**
- **Install Plugins.**



# Cisco Unified CallManager 5.0 Application Menu: IPMA

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Cisco IPMA Configuration Wizard

Cisco CM Attendant Console Plugins

Pilot Point

Cisco CM Attendant Console User

Cisco CM Attendant Console User File Upload

Cisco CallManager

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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Done 192.168.1.10

# Cisco Unified CallManager 5.0 Application Menu: Attendant Console

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Cisco IPMA Configuration Wizard

Cisco CM Attendant Console

Plugins

Pilot Point

Cisco CM Attendant Console User

Cisco CM Attendant Console User File Upload

Cisco CallManager

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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Done 192.168.1.10

# Cisco Unified CallManager 5.0 Application Menu: Plugins

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Cisco IPMA Configuration Wizard

Cisco CM Attendant Console

Plugins

Pilot Point

Cisco CM Attendant Console User

Cisco CM Attendant Console User File Upload

Cisco CallManager

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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Done 192.168.1.10

# Cisco Unified CallManager 5.0 User Management Menu

**The user management menu is used to configure user settings and rights:**

- **Users and rights for the Cisco Unified CallManager 5.0 system.**
- **Phone users for extension mobility and directory.**
- **Certification Authority Proxy Function (CAPF) for CTI, JTAPI and TAPI applications.**
- **SIP realm for SIP trunks to other systems.**

# Cisco Unified CallManager 5.0 User Management Menu: Users and Rights

The screenshot shows the Cisco CallManager Administration console in a Mozilla Firefox browser window. The address bar shows the URL `https://192.168.1.10/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The navigation menu includes: System, Call Routing, Media Resources, Voice Mail, Device, Application, **User Management**, Bulk Administration, Help, and Log Off.

The "User Management" dropdown menu is expanded, showing the following options:

- Application User
- End User
- Role
- User Group
- User/Phone Add
- Application User CAPF Profile
- End User CAPF Profile
- SIP Realm

The main content area features a "Cisco CallManager" logo and the following text:

System version: 5.0.1.51-414  
Administration version: 1.1.0.  
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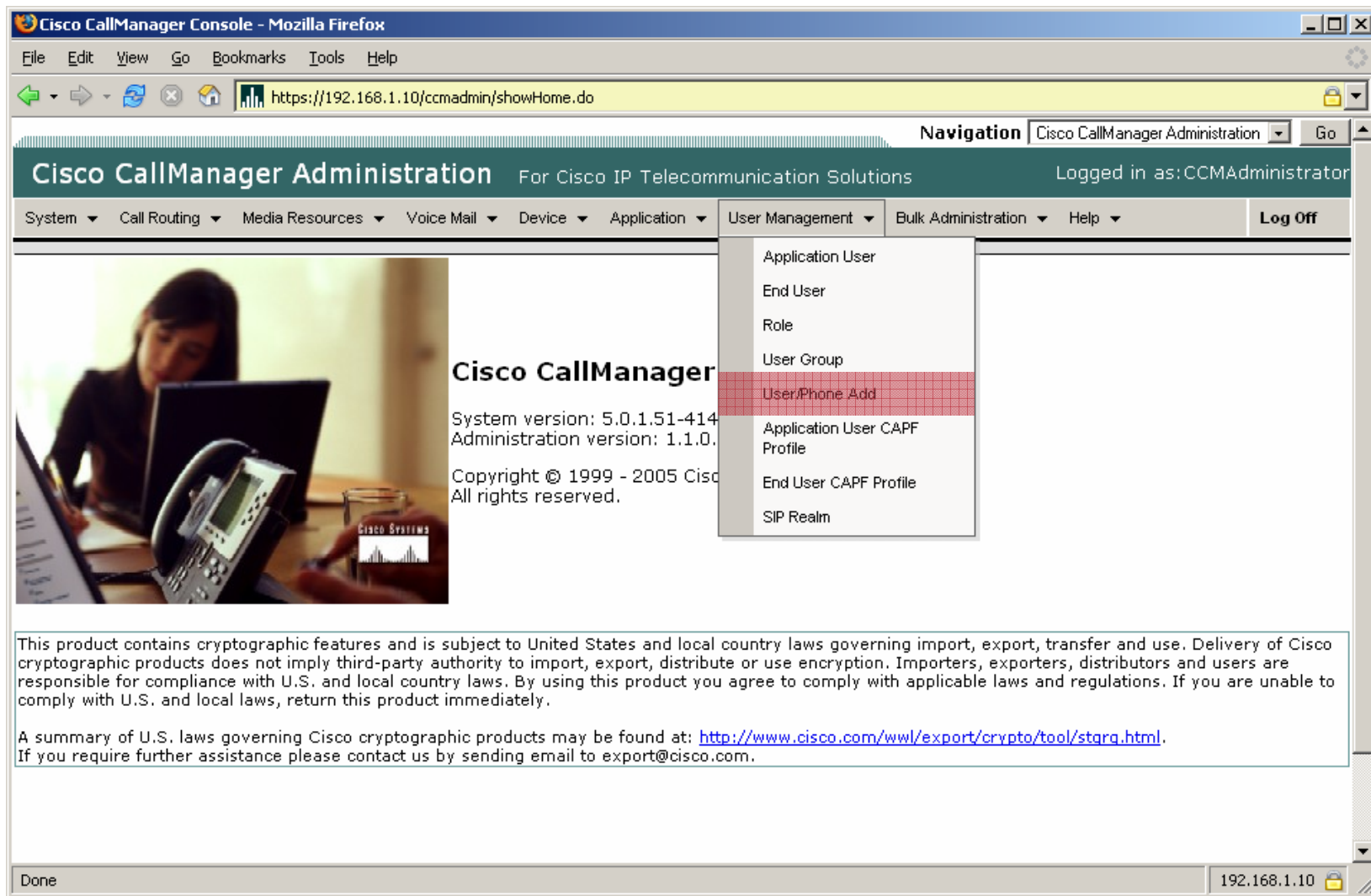
A legal disclaimer is present at the bottom of the page:

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If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 User Management Menu: Associate Users with Phones



The screenshot shows the Cisco CallManager Administration web interface. The browser window title is "Cisco CallManager Console - Mozilla Firefox". The address bar shows the URL "https://192.168.1.10/ccmadmin/showHome.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu is open, showing the following options: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, Help, and Log Off. The "User Management" menu is expanded, showing the following options: Application User, End User, Role, User Group, User/Phone Add (highlighted with a red grid pattern), Application User CAPF Profile, End User CAPF Profile, and SIP Realm. The main content area displays the Cisco CallManager logo, system version (5.0.1.51-414), administration version (1.1.0), and copyright information (© 1999 - 2005 Cisco Systems, Inc. All rights reserved.). A disclaimer and a link to U.S. laws governing Cisco cryptographic products are also visible.

**Cisco CallManager**  
System version: 5.0.1.51-414  
Administration version: 1.1.0.  
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If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

# Cisco Unified CallManager 5.0 User Management Menu: CAPF



The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser window. The browser's address bar displays the URL `https://192.168.1.10/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "User Management" menu is expanded, showing options: "Application User", "End User", "Role", "User Group", "User/Phone Add", "Application User CAPF Profile", "End User CAPF Profile", and "SIP Realm". The "Application User CAPF Profile" and "End User CAPF Profile" options are highlighted with a red grid pattern. The main content area features a "Cisco CallManager" logo, system version information (5.0.1.51-414), administration version (1.1.0), and copyright notice (© 1999 - 2005 Cisco Systems, Inc.). A legal disclaimer and a link to U.S. laws governing cryptographic products are also present. The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 User Management Menu: SIP Realm

The screenshot shows the Cisco CallManager Administration web interface. The browser window title is "Cisco CallManager Console - Mozilla Firefox". The address bar shows the URL "https://192.168.1.10/ccmadmin/showHome.do". The page header includes "Cisco CallManager Administration For Cisco IP Telecommunication Solutions" and "Logged in as: CCMAdministrator". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", "Help", and "Log Off". The "User Management" menu is open, displaying the following options: Application User, End User, Role, User Group, User/Phone Add, Application User CAPF Profile, End User CAPF Profile, and SIP Realm. The "SIP Realm" option is highlighted with a red grid pattern. The main content area features a photograph of a woman working at a desk with a laptop and a Cisco IP phone. Text on the page includes "Cisco CallManager", "System version: 5.0.1.51-414", "Administration version: 1.1.0.", and "Copyright © 1999 - 2005 Cisco Systems, Inc. All rights reserved.". A legal disclaimer is present at the bottom of the page, stating: "This product contains cryptographic features and is subject to United States and local country laws governing import, export, transfer and use. Delivery of Cisco cryptographic products does not imply third-party authority to import, export, distribute or use encryption. Importers, exporters, distributors and users are responsible for compliance with U.S. and local country laws. By using this product you agree to comply with applicable laws and regulations. If you are unable to comply with U.S. and local laws, return this product immediately. A summary of U.S. laws governing Cisco cryptographic products may be found at: <http://www.cisco.com/wwl/export/crypto/tool/stqrg.html>. If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com)." The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".



# Multi Level Admin

**Cisco Unified CallManager 5.0 configuration menu can be restricted using Multi Level Admin (MLA) feature:**

- **Groups can be used to give read or read/write access to menu and sub-menu content.**
- **Users can be assigned to groups to work with configured group rights.**

# Multi Level Admin

**To add a new user for MLA, the following tasks need to be performed:**

- **Add a new application user**
- **Add the new user to the user group(s)**

# Multi Level Admin

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser title is "Find and List Application Users - Mozilla Firefox". The address bar shows the URL: `https://192.168.1.10/ccadmin/appuserFindList.do?lookup=false&multiple=true&recCnt=12&colCnt=3`. The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The main content area is titled "Find and List Application Users" and contains a "+" icon, a "Status" section with "0 records found", a "Search Options" section with a search form, and a "Search Results" section with the message "No active query. Please enter your search criteria using the options above." The "Add New" button is highlighted with a red box, and a yellow callout bubble with the text "Add a new application user" and a circled "1" points to it.

# Multi Level Admin

**Save configuration** ③

**Configure UserID and password** ②

**Status**  
Status: Ready

**Application User Information**  
User ID \* SysAdmin  
Password \*  
Confirm Password \*  
Digest Credentials  
Confirm Digest Credentials  
Presence Group CHI-Phones

Accept Presence Subscription  
 Accept Out-of-dialog REFER  
 Accept Unsolicited Notification  
 Accept Replaces Header

**Device Information**  
Available Devices SEP000173AF21BE  
SEP000173AF21BF  
SEP000173AF21C1  
Find more Phones  
Find more Route Points

# Multi Level Admin

Find and List User Groups - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/userGroupFindList.do?lookup=false&multiple=true&recCnt=0&colCnt=3

Find and List User Groups

Search for user groups (4)

Select group to which user should be added (5)

Find Search Within Results

Search Results

Name	Roles	Copy
<a href="#">Standard CCM Admin Users</a>		
<a href="#">Standard CCM End Users</a>		
<a href="#">Standard CCM Gateway Administration</a>		
<a href="#">Standard CCM Phone Administration</a>		
<a href="#">Standard CCM Read Only</a>		
<a href="#">Standard CCM Server Maintenance</a>		
<a href="#">Standard CCM Server Monitoring</a>		
<a href="#">Standard CCM Super Users</a>		
<a href="#">Standard CTI Allow Call Park Monitoring</a>		
<a href="#">Standard CTI Allow Calling Number Modification</a>		
<a href="#">Standard CTI Allow Control of All Devices</a>		
<a href="#">Standard CTI Allow Reception of SRTP Key Material</a>		
<a href="#">Standard CTI Enabled</a>		
<a href="#">Standard CTI Secure Connection</a>		
<a href="#">Standard EM Authentication Proxy Rights</a>		
<a href="#">Standard Packet Sniffer Users</a>		

https://192.168.1.10/ccmadmin/userGroupEdit.do?key=40ca894c-cdf0-a1f9-f73a-8222240c67a1 192.168.1.10

# Multi Level Admin

User Group Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/userGroupEdit.do?key=40ca894c-cdf0-a1f9-f73a-8222240c67a1

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

User Group Configuration Related Links: Back To Find/List Go

Status  
Status: Ready

User Group Information  
Name\* Standard CCM Admin Users

Users in Group

	Full Name	User ID	Permission
<input type="checkbox"/>	<a href="#">TestViewAdmin</a>		

Add End Users to Group Add Application Users to Group Select All Clear All Delete Selected Rows per Page 50

Copy Add New

\*- indicates required item.

**Add new user to group** 6

Done 192.168.1.10

# Multi Level Admin

The screenshot shows a web browser window titled "Find and List Application Users" with the URL "https://192.168.1.10". The interface includes a search bar with a dropdown menu set to "begins with" and a "Find" button. Below the search bar is a table of search results:

	User ID	Copy
<input checked="" type="checkbox"/>	SysAdmin	
<input type="checkbox"/>	TestAdmin	

Below the table are buttons for "Select All", "Clear All", "Add Selected", and "Close". A "Rows per Page" dropdown is set to "50".

Callout 7: "Select users to add" points to the "Add Selected" button.

Callout 8: "Add user(s)" points to the "Add Selected" button.

# Summary

- **Cisco Unified CallManager 5.0 is an “out of the box” solution that can be administrated via a web interface.**
- **To access the Cisco Unified CallManager 5.0 administration page enter the IP of the system into your browser.**
- **The Cisco Unified CallManager 5.0 system menu is used to configure system and cluster settings.**
- **The call routing menu is used to configure the dial plan throughout the Cisco Unified CallManager 5.0 cluster.**
- **The media resources menu is used to manage Cisco Unified CallManager 5.0 media resources.**
- **The voice mail menu is used to configure the voice mail integration on Cisco Unified CallManager 5.0.**



## Summary (cont.)

- **The device menu is used to manage devices on Cisco Unified CallManager 5.0.**
- **The application menu is used to configure and install Cisco Unified CallManager 5.0 applications.**
- **The user management menu is used to manage permissions and rights on Cisco Unified CallManager 5.0.**
- **MLA allows to manage access to Cisco Unified CallManager 5.0 web pages.**

# CISCO SYSTEMS





## **Administering Cisco Unified CallManager 5.0**

# **Configuring CallManager 5.0 Basic Settings**

# Objectives

- **Basic Settings Configuration Overview**
- **DNS vs. IP**
- **Service Activation**
- **Service Parameters**
- **Phone NTP Reference**
- **Time Zone**
- **Enterprise Parameters**

# Basic Settings Configuration Overview

**Cisco Unified CallManager 5.0 should have a basic configuration before any device is configured:**

- **DNS vs. IP considerations should be made.**
- **Cisco Unified CallManager, TFTP server and CTI manager services need to be enabled.**
- **Service and enterprise parameters should be configured to meet the customers' criteria.**
- **Phone NTP Reference should be added to fully support SIP phones.**
- **At least the default time zone should be configured in order to use the correct local time and date.**

# DNS vs. IP

**Cisco Unified CallManager 5.0 can use DNS names or IP addresses for system address values.**

Advantages of IP addresses	Advantages of DNS
<ul style="list-style-type: none"><li>• <b>No need of DNS server.</b></li><li>• <b>Reduces potential source of failure.</b></li><li>• <b>Simplifies troubleshooting.</b></li></ul>	<ul style="list-style-type: none"><li>• <b>Simplified management because of names instead of numbers.</b></li><li>• <b>Easier IP address changes because of name based IP paths.</b></li></ul>

# Service Activation

**To enable Cisco Unified CallManager 5.0 services perform the following tasks:**

- **Access the Cisco Unified CallManager Serviceability.**
- **Go to “Tools” > “Service Activation”.**
- **Select your server.**
- **Enable the necessary services.**
- **Go to “Tools” > “Control Center – Feature Services” and select your server.**
- **Verify that the configured services are up and running.**

# Service Activation

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/showHome.do

Navigation Cisco CallManager Serviceability Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

**Cisco CallManager Administration**

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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Select "Cisco Unified CallManager Serviceability" and click "Go"

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If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

Done 192.168.1.10



# Service Activation

**Go to "Service Activation"**

Cisco CallManager Serviceability For Cisco IP Telecommunication Solutions

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1  
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If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

https://192.168.1.10/ccmservice/serviceactivation.jsp;jsessionid=7B230237E68A6ECD286743F2DAEE82CD 192.168.1.10

# Service Activation

The screenshot displays the Cisco CallManager Serviceability interface in a Mozilla Firefox browser window. The browser's address bar shows the URL: `https://192.168.1.10/ccmservice/serviceactivation.jsp;jsessionid=7B230237E68A6ECD286743F2DAEE82CD`. The page title is "Cisco CallManager Serviceability" with the subtitle "For Cisco IP Telecommunication Solutions". A navigation bar includes "Alarm", "Trace", "Tools", "Snmp", and "Help". The main content area is titled "Service Activation" and contains a "Select Server" section. A dropdown menu is open, showing the selected option "192.168.1.10" and other options like "- Select a Server -". A yellow callout box with a black border and a pointer to the dropdown contains the text: "Select the appropriate server." The browser's status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Service Activation

The screenshot shows the Cisco CallManager Serviceability - Service Activation page in a Mozilla Firefox browser. The page title is "Service Activation" and the URL is "https://192.168.1.10/ccmservice/serviceactivation.jsp?htxtNodeID=8fb74274-671f-491a-93e3-afb5ee74ed738&htxtSubmit=Status&publisherNode=tru".

Annotations and actions:

- Refresh page:** A callout points to the refresh icon in the top navigation bar.
- Save and perform settings:** A callout points to the save icon in the top navigation bar.
- Reset to default for single server operation:** A callout points to the status information "Status : Ready".
- Select the services that should be activated:** A callout points to the checked checkboxes in the "CM Services" list.
- Deselect the services that should be deactivated:** A callout points to the unchecked checkboxes in the "CM Services" list.
- Configured status:** A callout points to the "Deactivated" status in the "CM Services" table.

Service Name	Activation Status
<input checked="" type="checkbox"/> Cisco CallManager	Activated
<input checked="" type="checkbox"/> Cisco Tftp	Activated
<input type="checkbox"/> Cisco M...	Deactivated
<input type="checkbox"/> Cisco IP Voice	Deactivated
<input checked="" type="checkbox"/> Cisco CTIMana	Activated
<input type="checkbox"/> Cisco CallMan	Deactivated
<input type="checkbox"/> Cisco Extension Mobility	Deactivated
<input type="checkbox"/> Cisco ...ode	Deactivated
<input type="checkbox"/> Cisco CallM...	Deactivated
<input type="checkbox"/> Cisco Dialed N...	Deactivated
<input type="checkbox"/> Cisco DHCP M...	Deactivated

# Control Center

Navigation: Cisco CallManager Serviceability

Service Activation

Control Center - Feature Services

Control Center - Network Services

Serviceability Reports Archive

CDR Management

Related Links: Control Center - Feature Services

Select Server: 192.168.1.10

Service Name	Activation Status
<input checked="" type="checkbox"/> Cisco CallManager	Activated
<input checked="" type="checkbox"/> Cisco Tftp	Activated
<input type="checkbox"/> Cisco Messaging Interface	Deactivated
<input type="checkbox"/> Cisco IP Voice Media Streaming App	Deactivated
<input checked="" type="checkbox"/> Cisco CTIManager	Activated
<input type="checkbox"/> Cisco CallManager Attendant Console Server	Deactivated
<input type="checkbox"/> Cisco Extension Mobility	Deactivated
<input type="checkbox"/> Cisco Extended Functions	Deactivated
<input type="checkbox"/> Cisco CallManager Cisco IP Phone Services	Deactivated
<input type="checkbox"/> Cisco Dialed Number Analyzer	Deactivated

# Control Center

The screenshot shows the Cisco CallManager Serviceability Control Center interface. At the top, there is a navigation bar with the text "Cisco CallManager Serviceability For Cisco IP Telecommunication Solutions". Below this, there are several tabs and a "Control Center - Feature Services" section. A "Status" section contains three icons: a green circle, a red circle, and a blue circle with a refresh symbol. A yellow callout box points to these icons with the text "Stop, start, restart selected service".

Below the status icons is a "Select Server" section with a dropdown menu showing "Server: 192.168.1.10".

The main section is titled "CM Services" and contains a table with the following columns: "Service Name", "Status\*", and "Activation Status".

Service Name	Status*	Activation Status
<input checked="" type="radio"/> Cisco CallManager	Started	Activated
<input type="radio"/> Cisco TFTP	Started	Activated
<input type="radio"/> Cisco ...	Stopped	Deactivated
<input type="radio"/> Cisco ...	Stopped	Deactivated
<input type="radio"/> Cisco ...	Started	Activated
<input type="radio"/> Cisco CallManager Attendant Console Server	Stopped	Deactivated
<input type="radio"/> Cisco Extension Mobility	Stopped	Deactivated
<input type="radio"/> Cisco Extended Functions	Stopped	Deactivated
<input type="radio"/> Cisco CallManager Cisco IP Phone Services	Stopped	Deactivated
<input type="radio"/> Cisco Dialed Number Analyzer	Stopped	Deactivated

Yellow callout boxes provide additional context: "Select service to start, stop or restart" points to the radio button for "Cisco CallManager"; "Configured status" points to the "Activation Status" column; and "Actual status" points to the "Status\*" column.

# Service Parameters

**Service parameters on Cisco Unified CallManager 5.0 can be changed to fit special needs e.g.:**

- **T302 timer to speed up dialing.**
- **SIP retry counts to relieve overloaded SIP proxy servers.**
- **Enable call detail records.**
- **Enable MGCP gateway overlap receiving.**
- **Define extension mobility maximum login time.**
- **Define attendant console username.**
- **Change extended functions connecting port.**
- **Define voice media streaming application supported codecs.**

# Service Parameters

The screenshot shows the Cisco CallManager Administration interface for Service Parameter Configuration. The browser window title is "Service Parameter Configuration - Mozilla Firefox". The address bar shows the URL: `https://192.168.1.10/ccmadmin/serviceParamEdit.do?server=8fb74274-671f-491a-93e3-afb5ee74ed73&service=-1`. The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". A navigation menu is visible with options like System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The main content area is titled "Service Parameter Configuration" and includes a "Status" section showing "Status: Ready". Below this is the "Select Server and Service" section, which contains two dropdown menus. The "Server \*" dropdown is currently set to "192.168.1.10 (Active)". The "Service \*" dropdown is currently set to "- Not Selected -". A list of services is displayed below the "Service \*" dropdown, including "Cisco AMC Service (Inactive)", "Cisco Bulk Provisioning Service (Inactive)", "Cisco CAR Scheduler (Inactive)", "Cisco CTIManager (Active)", "Cisco CTL Provider (Active)", "Cisco CallManager (Active)", "Cisco CallManager Attendant Console Server (Inactive)", "Cisco CallManager SNMP Service (Inactive)", "Cisco Certificate Authority Proxy Function (Active)", "Cisco DRF Local (Active)", "Cisco DRF Master (Active)", "Cisco Database Layer Monitor (Active)", "Cisco DirSync (Inactive)", "Cisco Extended Functions (Inactive)", "Cisco Extension Mobility (Inactive)", "Cisco IP Manager Assistant (Inactive)", "Cisco IP Voice Media Streaming App (Inactive)", "Cisco Log Partition Monitoring Tool (Active)", and "Cisco Messaging Interface (Inactive)". Two yellow callout boxes with black text are present: "Select Server" points to the "Server \*" dropdown, and "Select Service" points to the "Service \*" dropdown. The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# Service Parameters

**Service Parameter Configuration - Mozilla Firefox**

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccadmin/serviceParamEdit.do?server=8fb74274-671f-491a-93e3-afb5ee74ed73&service=0

Service Parameter Configuration Related Links: Parameters for All Servers Go

**Status**  
Status: Ready

**Select Server and Service**  
Server \*: 192.168.1.10 (Active)  
Service \*: Cisco CallManager (Active)

All parameters apply only to the current server except parameters that are in the Clusterwide group(s).

**Cisco CallManager (Active) Parameters on server 192.168.1.10 (Active)**

Parameter Name	Parameter Value	Suggested Value
<b>CCM Call Throttling</b>		
<a href="#">Code Yellow Entry Latency</a> *	20	20
<a href="#">Code Yellow Exit Latency Calculation</a> *	40	40
<a href="#">Code Yellow Duration</a> *	99999	99999
<a href="#">Max Events Allowed</a> *	2000	2000
<a href="#">System Throttle Sample Size</a> *	10	10
<b>System</b>		
<a href="#">CDR Enabled Flag</a> *	False	False
<a href="#">CDR Log Calls with Zero Duration Flag</a> *	False	False
<a href="#">Night Analysis Complexity</a> *		StandardAnalysis

Done 192.168.1.10

**Verify/configure service settings**



# Phone NTP Reference

**In Cisco Unified CallManager 5.0 SIP phones can have NTP references using the following options:**

- **Directed broadcast:**
  - The phone can use information from any NTP server but gives the listed NTP servers priority.
- **Unicast:**
  - The phone will use information from configured NTP server.
  - If NTP server is unavailable, the phone will use Cisco Unified CallManager SIP registration information.

# Phone NTP Reference

## **NTP server recommendations:**

- **If phones do not have access to the internet make sure that there are local NTP server(s) that can be used.**
- **If phones are able to access the internet, an alternative external NTP server can be configured.**
- **If there is no local NTP server an external NTP server on the Internet should be used.**
- **External NTP servers could be e.g.:**
  - **pool.ntp.org**
  - **ntp2.usno.navy.mil**
  - **ntp.colby.edu**
  - **ntp.alaska.edu**

# Phone NTP Reference

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10/ccmadmin/ntpServerFindList.do`. The page title is "Find and List Phone NTP References - Mozilla Firefox". The navigation bar includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The main content area is titled "Find and List Phone" and contains a "+" icon, a search form, and an "Add New" button. Three yellow callout boxes provide instructions: one points to the "+" icon, another points to the search form, and a third points to the "Add New" button.

**Add a new NTP Reference**

**Search for existing NTP references**

**Add a new NTP Reference**

# Phone NTP Reference

The screenshot shows the Cisco CallManager Administration interface for configuring a Phone NTP Reference. The browser window title is "Phone NTP Reference Configuration - Mozilla Firefox" and the URL is "https://192.168.1.10/ccmadmin/ntpServerEdit.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The main content area is titled "Phone NTP Reference Configuration" and includes a "Status" section showing "Ready". The "Phone NTP Reference Information" section contains the following fields:

- IP Address: 192.168.3.254
- Description: Global NTP Server
- Mode\*: Directed Broadcast (selected from a dropdown menu)

The dropdown menu for Mode\* shows the following options: Directed Broadcast, Unicast, Multicast, Anycast, and Directed Broadcast. A "Save" button is located below the Mode\* dropdown. A callout box points to the IP Address and Description fields with the text "Enter IP address and description". Another callout box points to the Mode\* dropdown with the text "Select phone NTP mode". A red note at the bottom of the screenshot states: "Note: Currently only Broadcast and Unicast are supported".

# Time Zone

**Time groups on Cisco Unified CallManager 5.0 allow to:**

- **Map devices to correct time zone in order to use correct local time.**
- **Define the time format to meet regional criteria.**
- **Define NTP servers that should be used by SIP phones.**

# Time Zone

**Find and List Date/Time Groups**

Status  
1 records found

**Search Options**  
Find Date/Time Group where Group Name begins with Find  Search Within Results  
(datetimesetting.name begins with any)

**Search Results**

Name	Time Zone
<a href="#">CMLocal</a>	Greenwich Standard Time

Add New Select All Clear All Delete Selected Rows per Page 50

**Search for existing Date/Time Groups**

**Select default time zone CMLocal**

# Time Zone

The screenshot shows the Cisco CallManager Administration interface for configuring a Date/Time Group. The page title is "Date/Time Group Configuration" and the user is logged in as "CCMAdministrator". The configuration details for the "CMLocal" group are as follows:

- Status:** Ready
- Date/Time Group Information:**
  - Date/Time Group: CMLocal (used by 7 devices)
  - Group Name: CMLocal
  - Time Zone: Pacific Standard/Daylight Time - (GMT-08:00) Pacific Time (US & Canada); Tijuana
  - Separator: / (slash) (applies to Date Format only)
  - Date Format: M/D/Y
  - Time Format: 12-hour
- Selected Phone NTP References:** (Ordered by highest priority)
- Buttons:** Add Phone NTP References, Remove Phone NTP References
- Footer Buttons:** Save, Delete, Copy, Reset, Add New

Three callout boxes provide instructions:

- Configure time and date format:** Points to the Date Format and Time Format dropdowns.
- Select proper time zone:** Points to the Time Zone dropdown.
- Add NTP Reference for phones within that group:** Points to the "Add Phone NTP References" button.

# Time Zone

The screenshot displays the Cisco CallManager Administration web interface. The main window is titled "Date/Time Group Configuration" and shows a search for NTP references. A secondary window, "Find and List Phone NTP References", is open, displaying search results for IP addresses. The search criteria are "IP Address" and "begins with". The search results table shows one entry: "192.168.3.254" with the description "Global NTP Server". The "Add Selected" button is highlighted, and a yellow callout box points to it with the text "Add selected NTP to date/time group". Another yellow callout box points to the search criteria with the text "Select NTP server(s) that should be used".

**Select NTP server(s) that should be used**

IP Address	Description
<input checked="" type="checkbox"/> 192.168.3.254	Global NTP Server

**Add selected NTP to date/time group**



# Time Zone

The screenshot shows the Cisco CallManager Administration interface for configuring a Date/Time Group. The browser window title is "Date/Time Group Configuration - Mozilla Firefox". The address bar shows the URL: `https://192.168.1.10/ccmadmin/datetimeGroupEdit.do?key=9ec4850a-7748-11d3-bdf0-00108302ead1`. The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator".

Two callouts are present:

- A yellow callout box with the text "Reset group to make changes become effect" points to the "Reset" button in the bottom toolbar.
- A yellow callout box with the text "Save changes" points to the "Save" button in the bottom toolbar.

The main configuration area is titled "Date/Time Group Configuration" and includes the following fields:

- Group Name \***: CMLocal
- Time Zone \***: Pacific Standard/Daylight Time - (GMT-08:00) Pacific Time (US & Canada); Tijuana
- Separator \***: / (slash) (applies to Date Format only)
- Date Format \***: M/D/Y
- Time Format \***: 12-hour
- Selected Phone NTP References (Ordered by highest priority)**: 192.168.3.254

Buttons at the bottom include "Save", "Delete", "Copy", "Reset", and "Add New". The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Enterprise Parameters

**Enterprise parameters are used to define cluster wide system settings like:**

- **Default phone protocol (SCCP or SIP).**
- **Cisco Unified CallManager user web-page content.**
- **System URLs used by Phones.**

# Enterprise Parameters

The screenshot displays the Cisco CallManager Administration web interface. The browser title is "Enterprise Parameters Configuration - Mozilla Firefox". The address bar shows the URL: <https://192.168.1.10/ccmadmin/serviceParamEdit.do?service=11>. The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator".

The main content area is titled "Enterprise Parameters Configuration" and shows a table of parameters. The "Auto Registration Phone Protocol" parameter is highlighted with a red box, and its dropdown menu is open, showing the following options:

Parameter Name	Parameter Value	Suggested Value
<a href="#">Synchronization Between Auto Device Profile and Phone Configuration</a> *	True	
<a href="#">Max Number of Device Level Trace</a> *	12	
<a href="#">DSCP for Phone-based Services</a> *	default DSCP (000000)	default DSCP (000000)
<a href="#">DSCP for Phone Configuration</a> *	CS3(precedence 3) DSCP (011000)	CS3(precedence 3) DSCP (011000)
<a href="#">DSCP for Cisco CallManager to Device Interface</a> *	CS3(precedence 3) DSCP (011000)	CS3(precedence 3) DSCP (011000)
<a href="#">Connection Monitor Duration</a> *	120	120
<a href="#">Auto Registration Phone Protocol</a> *	SCCP	SCCP
<a href="#">BLF For Call Lists</a> *	Disabled	Disabled
<a href="#">TFTP Encrypted Configuration</a> *	False	False

A yellow callout box with the text "Select protocol for auto-registration" points to the "Auto Registration Phone Protocol" dropdown menu.

# Enterprise Parameters

**Enterprise Parameters Configuration - Mozilla Firefox**

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/serviceParamEdit.do?s

**Select settings that should be available to phone users**

<a href="#">Max List Box Items</a> *	250	
<a href="#">Max Lookup Items</a> *	1000	1000
<a href="#">Enable Dependency Records</a> *	False	False

**CCMUser Parameters**

<a href="#">Show Ring Settings</a> *	False	False
<a href="#">Show Call Forwarding</a> *	True	True
<a href="#">Show Speed Dial Settings</a> *	True	True
<a href="#">Show Cisco IP Phone Services Settings</a> *	True	True
<a href="#">Show Personal Address Book Settings</a> *	False	True
<a href="#">Show Message Waiting Lamp Policy Settings</a> *	True	True
<a href="#">Show Line Text Label Settings</a> *	False	False
<a href="#">Show Locale for Phone Settings</a> *	True	True
<a href="#">Show Locale for Web Pages Settings</a> *	True	True
<a href="#">Show Change Password Option</a> *	True	True
<a href="#">Show Change PIN Option</a> *	True	True
<a href="#">Show Download Plugin Option</a> *	False	True
<a href="#">Show Online Guide Option</a> *	False	True

**CDR Parameters**

<a href="#">CDR File Time Interval</a> *	1	1
--	---	---

Done 192.168.1.10

# Enterprise Parameters

Enterprise Parameters Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccadmin/serviceParamEdit.do?service=11

[Precedence Alternate Party Timeout](#)\* 30 30

[Use Standard VM Handling For Precedence Calls](#)\* False

**Security Parameters**

[Cluster Security Mode](#)\* 0

[CAPF Phone Port](#)\* 3804 3804

[CAPF Operation Expires in \(days\)](#)\* 10 10

**Phone URL Parameters**

[URL Authentication](#) http://192.168.1.10:8080/ccmcip/authenticate.jsp

[URL Directories](#) http://192.168.1.10:8080/ccmcip/xmldirectory.jsp

[URL Idle](#)

[URL Idle Time](#) 0 0

[URL Information](#) http://192.168.1.10:8080/ccmcip/GetTelecasterHelpText.jsp

[URL Messages](#)

[IP Phone Proxy Address](#)

[URL Services](#) http://CM1:8080/ccmcip/getservicesmenu.jsp

**User Search Parameters**

[Enable All User Search](#)\* True True

[User Search Limit](#)\* 64 64

**CCM Web Services Parameters**

[Allowed Performance Oqueries Per Minute](#) 50 50

Done 192.168.1.10

**Change the name to an IP address if DNS services are not used**

# Summary

- **Cisco Unified CallManager 5.0 needs a basic configuration before any device is being added.**
- **Use IP addresses for Cisco Unified CallManager 5.0 to avoid dependencies on DNS servers.**
- **Services are activated through the service activation within the Cisco Unified CallManager serviceability.**
- **Service parameters are used to configure Cisco Unified CallManager services to apply individual needs.**
- **Time zones guarantee the correct time settings throughout multi-site networks.**
- **Enterprise parameters are used to define cluster wide parameter values.**

# CISCO SYSTEMS





## **Administering Cisco Unified CallManager 5.0**

# **Serviceability and Platform Administration**



# Objectives

- **Serviceability and Platform Administration Overview**
- **Cisco Unified CallManager 5.0 Alarms**
- **Cisco Unified CallManager 5.0 Traces**
- **Tools Menu**
- **SNMP Configuration**
- **CLI Commands and Platform Administration**
- **Additional Tools to Manage Cisco Unified CallManager 5.0**

# Serviceability and Platform Administration Overview

**Cisco Unified CallManager system and platform management are done via the web, CLI, and management tools:**

- **Cisco Unified CallManager Serviceability allows to configure necessary alarms and traces for Cisco Unified CallManager 5.0.**
- **Cisco Unified CallManager Platform Administration is used to manage the Cisco Unified CallManager 5.0 Linux platform.**
- **Cisco Unified CallManager SSH access allows to manage the system using CLI commands.**
- **Cisco Unified CallManager 5.0 can be monitored with Network Management applications.**

# Cisco Unified CallManager 5.0 Alarms

## Alarms on Cisco Unified CallManager 5.0:

- Alarms are configured via the Cisco Unified CallManager Serviceability.
- Alarms can be defined for each service individually.
- Alarms can be stored locally on the system and/or a syslog server.
- The alarm level can be defined individually for every destination.
- Use alarms to pinpoint problem areas in your network.

# Cisco Unified CallManager 5.0 Alarms

**Configure alarm settings**

**View alarm definitions**

**Access the Serviceability page.**

Administration version: 1.1.0.0-1  
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If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

# Cisco Unified CallManager 5.0 Alarms

The screenshot shows the Cisco CallManager Serviceability Alarm Configuration page in a Mozilla Firefox browser. The page title is "Cisco CallManager Serviceability - Alarm Configuration". The browser address bar shows the URL: <https://192.168.1.10/ccmservice/alarmconfig.jsp;jsessionid=10DC758B9B04212B13DE009AA97AB165?NodeID=8fb74274-671f-491a-93e3-afb5ee74ec>. The page has a navigation bar with "Cisco CallManager Serviceability" and a "Go" button. Below the navigation bar is a menu with "Alarm", "Trace", "Tools", "Snmp", and "Help". The main content area is titled "Alarm Configuration" and contains a "Select Server and Service" section. The "Server" dropdown menu is set to "192.168.1.10" and is highlighted with a yellow callout box labeled "1" with the text "Select server for which alarms should be defined". The "Service" dropdown menu is open, showing a list of services, and is highlighted with a yellow callout box labeled "2" with the text "Select service for which alarms should be defined". The services listed are: Cisco Bulk Provisioning Service (Inactive), Cisco CDR Agent (Active), Cisco CDR Repository Manager (Active), Cisco CTI Manager (Active), Cisco CallManager (Active), Cisco CallManager Attendant Console Server (Inactive), Cisco DHCP Monitor Service (Inactive), Cisco DRF Local (Active), Cisco DRF Master (Active), Cisco Database Layer Monitor (Active), Cisco DirSync (Inactive), Cisco Extended Functions (Inactive), Cisco IP Voice Media Streaming App (Inactive), Cisco License Manager (Active), Cisco Messaging Interface (Inactive), Cisco RIS Data Collector (Active), Cisco Tftp (Active), Cisco Tomcat (Active), and Cisco Trace Collection Service (Active). The status of each service is indicated in parentheses. The browser status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 Alarms

The screenshot shows the Cisco CallManager Serviceability Alarm Configuration page in a Mozilla Firefox browser. The page title is "Cisco CallManager Serviceability - Alarm Configuration - Mozilla Firefox". The URL is "https://192.168.1.10/ccmservice/alarmconfig.jsp;jsessionid=10DC758B9B04212B13DE009AA97AB165?NodeID=8fb74274-671f-491a-93e3-afb5ee74ec". The page has a navigation bar with "Cisco CallManager Serviceability" and a "Go" button. Below the navigation bar is a menu with "Alarm", "Trace", "Tools", "Snmp", and "Help". The main content area is titled "Alarm Configuration" and contains several sections:

- Status:** Shows "Status : Ready". A callout bubble with the number 3 points to a dropdown menu with the text "Enable/disable alarms".
- Local Syslogs:** Contains a checkbox "Enable Alarm" which is checked and highlighted with a red box. A callout bubble with the number 4 points to a dropdown menu for "Alarm Event Level" with the text "Select alarm level for each logging destination individually". The dropdown menu is open, showing options: Error, Emergency, Alert, Critical, Error (highlighted), Warning, Notice, Informational, and Debug.
- Remote Syslogs:** Contains a checkbox "Enable Alarm" which is unchecked. It also has a "Server Name \*" field.
- SDI Trace:** Contains a checkbox "Enable Alarm" which is checked.
- SDL Trace:** Contains a checkbox "Enable Alarm" which is checked.

At the bottom of the page, there is a "Done" button and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 Alarms

## Alarm Event Levels:

Level	Name	Description
7	Emergency	System is unusable.
6	Alert	Immediate action is needed.
5	Critical	Critical condition is detected.
4	Error	Error condition exists.
3	Warning	Warning condition is detected.
2	Notice	Normal but significant condition.
1	Informational	Information messages only.
0	Debug	Detailed event information used for debugging by Cisco TAC engineers.

# Cisco Unified CallManager 5.0 Traces

## Traces on Cisco Unified CallManager 5.0:

- Traces are configured via the Cisco Unified CallManager Serviceability.
- Traces can be defined for each Cisco Unified CallManager server individually.
- Tracing can be enabled/disabled for each parameter of a service individually.
- Traces are stored locally on the system.



# Cisco Unified CallManager 5.0 Traces

**Go to trace configuration** ①

**Cisco CallManager Serviceability**

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

javascript:void(0) 192.168.1.10

# Cisco Unified CallManager 5.0 Traces

The screenshot shows the Cisco CallManager Serviceability Trace Configuration page in a Mozilla Firefox browser. The page title is "Cisco CallManager Serviceability - Trace Configuration - Mozilla Firefox". The browser address bar shows the URL: <https://192.168.1.10/ccmservice/traceconfig.jsp;jsessionid=10DC758B9B04212B13DE009AA97AB165?NodeID=8fb74274-671f-491a-93e3-afb5ee74ec>. The page has a navigation bar with "Cisco CallManager Serviceability" and a "Go" button. Below the navigation bar is a menu with "Alarm", "Trace", "Tools", "Snmp", and "Help". The main content area is titled "Trace Configuration" and contains a "Select Server and Service" section. The "Server" dropdown menu is set to "192.168.1.10". The "Service" dropdown menu is open, showing a list of services with their status (Active or Inactive). A yellow callout box with a circled "2" points to the "Server" dropdown and contains the text "Select server for which tracing should be defined". Another yellow callout box with a circled "3" points to the "Service" dropdown and contains the text "Select service for which tracing should be defined". The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

**Select server for which tracing should be defined** ②

**Select service for which tracing should be defined** ③

Server: 192.168.1.10

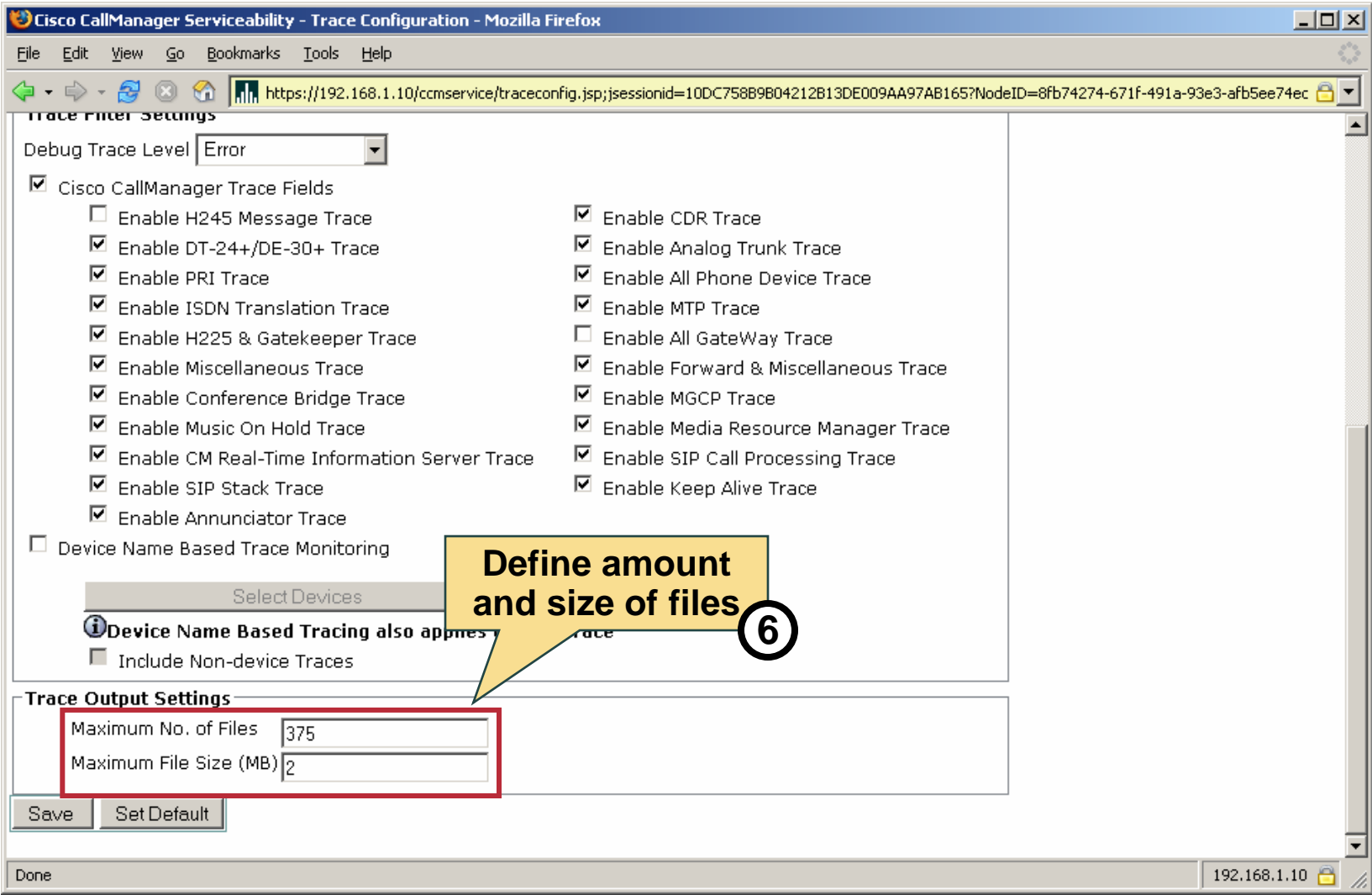
Service: - Select a Service -

- Select a Service -
- Cisco AMC Service (Inactive)
- Cisco AXL Web Service (Inactive)
- Cisco Bulk Provisioning Service (Inactive)
- Cisco CAR Scheduler (Inactive)
- Cisco CAR Web Service (Inactive)
- Cisco CCM DBL Web Library (Active)
- Cisco CCM NCS Web Library (Active)
- Cisco CCM PD Web Service (Active)
- Cisco CCMAAdmin Web Service (Active)
- Cisco CCMRealm Web Service (Active)
- Cisco CCMService Web Service (Active)
- Cisco CDR Agent (Active)
- Cisco CDR Repository Manager (Active)
- Cisco CTI Manager (Active)
- Cisco CTL Provider (Active)
- Cisco CallManager (Active)
- Cisco CallManager Attendant Console Server (Inactive)
- Cisco CallManager Cisco IP Phone Services (Inactive)
- Cisco CallManager SNMP Service (Inactive)

# Cisco Unified CallManager 5.0 Traces

The screenshot shows the Cisco CallManager Serviceability Trace Configuration page in a Mozilla Firefox browser. The page title is "Cisco CallManager Serviceability - Trace Configuration - Mozilla Firefox". The address bar shows the URL: <https://192.168.1.10/ccmservice/traceconfig.jsp;jsessionid=10DC758B9B04212B13DE009AA97AB165?NodeID=8fb74274-671f-491a-93e3-afb5ee74ec>. The page has a navigation bar with "Cisco CallManager Serviceability" and a "Go" button. Below the navigation bar is a menu with "Alarm", "Trace", "Tools", "Snmp", and "Help". The main content area is titled "Trace Configuration" and includes a "Related Links" section with "SDL Configuration" and a "Go" button. The "Status" section shows "Status: Ready". The "Service" section has a "Trace On" checkbox checked, highlighted by a red box and a yellow callout bubble with the text "Enable/disable tracing" and the number "4". The "Trace Filter Settings" section includes a "Debug Trace Level" dropdown set to "Error". Under "Cisco CallManager Trace Fields", several checkboxes are checked, including "Enable CDR Trace", "Enable Analog Trunk Trace", "Enable All Phone Device Trace", "Enable MTP Trace", and "Enable H225 & Gatekeeper Trace". These checked checkboxes are highlighted by red boxes and a yellow callout bubble with the text "Select/deselect parameters that should be included in trace output" and the number "5". The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 Traces



# Cisco Unified CallManager 5.0 Traces

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmservice/

Navigation Cisco CallManager Serviceability Go

## Cisco CallManager Serviceability

Alarm Trace Tools Snmp Help

Configuration

Troubleshooting Trace Settings

**Go to troubleshooting trace settings** ①

### Cisco CallManager Serviceability

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

javascript:void(0) 192.168.1.10

# Cisco Unified CallManager 5.0 Traces

The screenshot shows the 'Cisco CallManager Serviceability - Troubleshooting Trace Settings' page in a Mozilla Firefox browser. The page title is 'Cisco CallManager Serviceability For Cisco IP Telecommunication Solutions'. The main content area is titled 'Troubleshooting Trace Settings' and contains a table with the following data:

Services	Select all Nodes for a Service	192.168.1.10
Check all Services for a Node	<input type="checkbox"/>	<input type="checkbox"/>
Cisco CallManager	<input checked="" type="checkbox"/>	<input checked="" type="checkbox"/>
Cisco Tftp	<input type="checkbox"/>	<input type="checkbox"/>
Cisco Messaging Interface	<input type="checkbox"/>	N/A
Cisco IP Voice Media Streaming App	<input type="checkbox"/>	N/A
Cisco CallManager Attendant Console Server	<input type="checkbox"/>	N/A
Cisco Database Layer Monitor	<input type="checkbox"/>	<input type="checkbox"/>
Cisco CTIManager	<input type="checkbox"/>	<input type="checkbox"/>
Cisco RIS Data Collector	<input type="checkbox"/>	<input type="checkbox"/>
Cisco Certificate Authority Proxy Function	<input type="checkbox"/>	<input type="checkbox"/>
Cisco Log Partition Monitoring Tool	<input type="checkbox"/>	<input type="checkbox"/>
Cisco CallManager SNMP Service	<input type="checkbox"/>	N/A
Cisco Extended Functions	<input type="checkbox"/>	N/A
Cisco Bulk Provisioning Service	<input type="checkbox"/>	N/A
Cisco AMC Service	<input type="checkbox"/>	N/A
Cisco CAR Scheduler	<input type="checkbox"/>	N/A

Callout 1: Select service for tracing (2) - points to the checked checkbox for Cisco CallManager.

Callout 2: Inactive services are shown as N/A - points to the N/A status of several services.

# Cisco Unified CallManager 5.0 Traces

The screenshot displays the Cisco CallManager Serviceability Trace Configuration interface. The browser window title is "Cisco CallManager Serviceability - Trace Configuration - Mozilla Firefox". The address bar shows the URL: <https://192.168.1.10/ccmservice/traceconfig.jsp;jsessionid=10DC758B9B04212B13DE009AA97AB165?NodeID=8fb74274-671f-491a-93e3-afb5ee74ec>. The page header includes "Cisco CallManager Serviceability" and "For Cisco IP Telecommunication Solutions". The main content area is titled "Trace Configuration" and includes a "Related Links" section with a dropdown menu set to "SDL Configuration". The "Status" section indicates that troubleshooting traces have been set for this service. The "Select Server and Service" section shows the server as "192.168.1.10" and the service as "Cisco CallManager (A)". The "Trace On" checkbox is checked and highlighted with a red box. The "Trace Filter Settings" section shows the "Debug Trace Level" set to "Detailed" and several checkboxes checked, including "Cisco CallManager Trace Fields", "Enable H245 Message Trace", "Enable DT-24+/DE-30+ Trace", "Enable PRI Trace", "Enable ISDN Translation Trace", "Enable H225 & Gatekeeper Trace", "Enable CDR Trace", "Enable Analog Trunk Trace", "Enable All Phone Device Trace", "Enable MTP Trace", and "Enable All GateWay Trace". A yellow callout box points to the "Trace On" checkbox with the text: "As soon as tracing is active in the 'troubleshooting trace settings' no changes can be made via the 'trace configuration'." The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Cisco Unified CallManager 5.0 Traces

Generic Log Viewer for service "Cisco CallManager" and trace type "sdi"

Enter a Search String  Search  Match case

File Content

```
02/21/2006 02:43:20.009 CCM|<--CLineStateNotifier:SendData() |<CLID::StandAloneCluster><NID::192.168.1.10><LVL::Entry_exit><MASK::0001>
02/21/2006 02:43:20.009 CCM|<--CLineStateNotifier:DelLine() |<CLID::StandAloneCluster><NID::192.168.1.10><LVL::Entry_exit><MASK::0001>
02/21/2006 02:43:20.009 CCM|StationD: (0000002) INFO restart0_CcCiRes: updating CI=32222298 for cdpc=6|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.010 CCM|LineControl(2): star_DSetCallState(1), State of cdpc (11) is 1|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.010 CCM|StationD: (0000002) DEBUG- star_DSetCallState(2) State of cdpc(6) is 1.|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.011 CCM|StationD: (0000002) SetLamp mode=2, stim=9 stimInst=1.|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.011 CCM|StationD: (0000002) DEBUG- star_DSetCallPhase updateACall=32222298 from Phase=0 to callPhase=0.|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.012 CCM|StationD: (0000002) CallState callState=1 lineInstance=1 callReference=32222298 privacy=0 precedenceLv=4 precedenceDrm=0|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.012 CCM|StationD: (0000002) DisplayPromptStatus timeOut=0 Status='1/2%' content='Enter Number' line=1 CI=32222298 ver=84000006.|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.012 CCM|StationD: (0000002) StationOutputDisplayText don't need to send, because mlsALegacyDevice = 0|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.012 CCM|StationD: (0000002) ActivateCallPlane lineInstance=1.|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.012 CCM|StationD: (0000002) DEBUG- star_DSetCallState(4) State of cdpc(6) is 2.|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.013 CCM|LineCdpc(11): -dispatchToAllDevices-, sigName=LmKeyReplaceReq, device=SEP000BDBDF61DE|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.015 CCM|Digit Analysis: wait_DaReq: Matching Legacy Numeric, digits=2302|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.016 CCM|Digit Analysis: getDaRes - voiceMailCallingSearchSpace=|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE>
02/21/2006 02:43:20.016 CCM|Digit analysis: match(pi="2", fqc="31 25553001", cn="3001", plv="5", pss="CHI-Phones:Intersite-Routing:CHI-Intl-PSTN:CHI-LD-PSTN:CHI-Local-PSTN:CHI-911", TodFilteredPss=|<CLID::StandAloneCluster><NID::192.168.1.10><CT::1,100,59,1.199><IP::10.128.128.134><DEV::SEP000BDBDF61DE><LVL::State Transition><MASK::0001>
02/21/2006 02:43:20.016 CCM|PretransformCallingPartyNumber=3001
|CallingPartyNumber=3001
|DialingPartition=CHI-Phones
|DialingPattern=2302
|FullyQualifiedCalledPartyNumber=2302
|DialingPatternRegularExpression=(2302)
|DialingWhere=
|PatternType=Enterprise
|PotentialMatches=NoPotentialMatchesExist
|DialingSdiProcessId=(0,0,0)
|PretransformDigitString=2302
|PretransformTagsList=SUBSCRIBER
|PretransformPositionalMatchList=2302
|CollectedDigits=2302
|UnconsumedDigits=
|TagsList=SUBSCRIBER
|PositionalMatchList=2302
|VoiceMailbox=
|VoiceMailCallingSearchSpace=
|VoiceMailPilotNumber=
```

3001 calls to 2302

Show New Data Clear Close



# Cisco Unified CallManager 5.0 Traces: Trace Collection Tool

## Significant Enhancements:

- Scheduled trace collection
- Download trace on Alert Central
- Real time view
- Real time events
- Search strings
- Download & delete
- Absolute & relative time query
- Generic queries
- SFTP push
- Job status screen

The screenshot displays the Cisco CallManager Serviceability Real-Time Monitoring Tool interface. The top window is the 'Collect Files' dialog, which allows users to specify collection time (Absolute or Relative Range) and download options (Partition, Directory, Zip Files, etc.). The bottom window is the 'Alert Central' screen, which displays a table of alert statuses and an 'Alert Detail' pop-up window showing a 'Critical Service Down' alert for the CAR Scheduler.

Alert Name	Enabled	In Safe Range	Alert Action	Last Alert Raised
BeginThrottlingCallListBLFSubscriptions	Enabled	Yes	Default	N/A
CallProcessingNodeCpuPegging	Enabled	Yes	Default	N/A
CDRAgentSendFileFailed	Enabled	N/A	Default	N/A
CDRFileDeliveryFailed			Default	N/A
CDRHighWaterMarkExceeded			Default	N/A
CDRMaximumDiskSpaceExceeded			Default	N/A
CodeYellow			Default	N/A
CriticalServiceDown			Default	04:27:07 PM 03/21/06
DBReplicationFailure			Default	N/A
ExcessiveVoiceQualityReports			Default	N/A
LogFileSearchStringFound			Default	N/A
LogPartitionHighWaterMarkExceeded			Default	N/A
LogPartitionLowWaterMarkExceeded			Default	N/A
LowActivePartitionAvailableDiskSpace			Default	N/A
LowAttendantConsoleServerHeartbeat			Default	N/A
LowAvailableVirtualMemory			Default	N/A
LowCallManagerHeartbeat			Default	N/A
LowInactivePartitionAvailableDiskSpace	Enabled	Yes	Default	N/A
LowSwapPartitionAvailableDiskSpace	Enabled	Yes	Default	N/A
LowTCPConnectionPoolSize			Default	N/A

**Alert Detail:**  
On Tue Mar 21 07:27:07 PST 2006 on node 192.168.1.10.  
Service status is DOWN.  
The following service(s) are down: Cisco CAR Scheduler.

# Tools Menu

**The Cisco Unified CallManager 5.0 serviceability tools menu contains the following tools:**

Name	Description
<b>Service Activation</b>	<b>Activate and deactivate Cisco Unified CallManager 5.0 services.</b>
<b>Control Center – Feature Services</b>	<b>View status and to start, stop, and restart Cisco Unified CallManager 5.0 feature services.</b>
<b>Control Center – Network Services</b>	<b>View status and to start, stop, and restart Cisco Unified CallManager 5.0 network services.</b>
<b>Serviceability Reports Archive</b>	<b>View reports generated by the serviceability reporter service.</b>
<b>CDR Management</b>	<b>Manage call detail record functionality on Cisco Unified CallManager 5.0.</b>

# Tools Menu

The screenshot shows the Cisco CallManager Serviceability web interface. The browser title is "Cisco CallManager Console - Mozilla Firefox" and the address bar shows "https://192.168.1.10/ccmservice/". The page header includes "Navigation" and "Cisco CallManager Serviceability". A menu bar contains "Alarm", "Trace", "Tools", "Snmp", and "Help". The "Tools" menu is expanded, listing the following items:

- Service Activation
- Control Center - Feature Services
- Control Center - Network Services
- Serviceability Reports Archive
- CDR Management

A yellow callout box with the text "Select tool" points to the "Tools" menu. The main content area displays system information: "System version: 5.0.1.51-414", "Administration version: 1.1.0.0-1", and "Copyright © 1999 - 2005 Cisco Systems, Inc. All rights reserved." Below this is a legal disclaimer regarding cryptographic features and a link to Cisco's export policy page.

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javascript:void(0) 192.168.1.10

# SNMP Configuration

## **Simple Network Management Protocol (SNMP) support on Cisco Unified CallManager 5.0:**

- **SNMP V1 and V2 including community string.**
- **SNMP V3 user.**
- **System group MIB-II table.**

# SNMP Configuration

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmservice/

Navigation Cisco CallManager Serviceability Go

## Cisco CallManager Serviceability

For Cisco IP Telecommunication Solutions

Alarm Trace Tools Snmp Help

- V1/V2c
- V3
- SystemGroup

**Configure SNMP settings**

**Cisco CallManager Serviceability**

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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javascript:void(0) 192.168.1.10

# SNMP Configuration

- The V1/V2c menu is used to configure SNMP community string and notification destination.
- The V3 sub-menu is used to configure SNMP V3 parameters.
- The system group sub-menu is used to configure the MIB-II system group system contact and system location objects.



# Platform Administration

There are two ways on Cisco Unified CallManager 5.0 to manage platform and system:

- Platform administration web GUI
- Platform administration CLI



Cisco IPT Platform Administration

Username and password are case sensitive

## Logon

Username	<input type="text" value="administrator"/>
Password	<input type="password" value="administrator"/>
<input type="button" value="Submit"/> <input type="button" value="Reset"/>	

```
192.168.1.10 - PuTTY
login as: administrator
administrator@192.168.1.10's password:

Welcome to the Platform Command Line
Interface (version 1.1)

admin: █
```

# Platform Administration

## CLI commands:

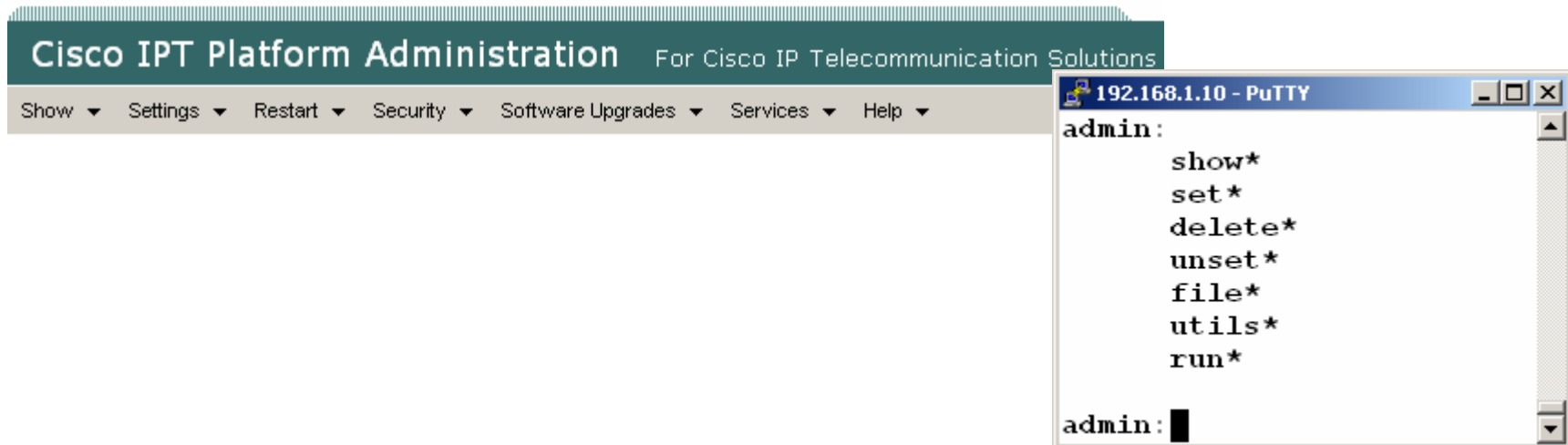
Name	Description
<b>show*</b>	<b>Show system parameters.</b>
<b>set*</b>	<b>Set system parameters and enable functionality.</b>
<b>delete*</b>	<b>Delete parameters.</b>
<b>unset*</b>	<b>Disable functionality.</b>
<b>file*</b>	<b>Manage files on the system.</b>
<b>utils*</b>	<b>Use system utilities.</b>
<b>run*</b>	<b>Start processes on the system.</b>



# Platform Administration

On the Platform administration web page most CLI commands can be used:

- Some are identical like web based ping and CLI utils network ping.
- Some are condensed on web page like network setting.



# Additional Tools to Manage Cisco Unified CallManager 5.0

## Real-Time Monitoring Tool (RTMT):

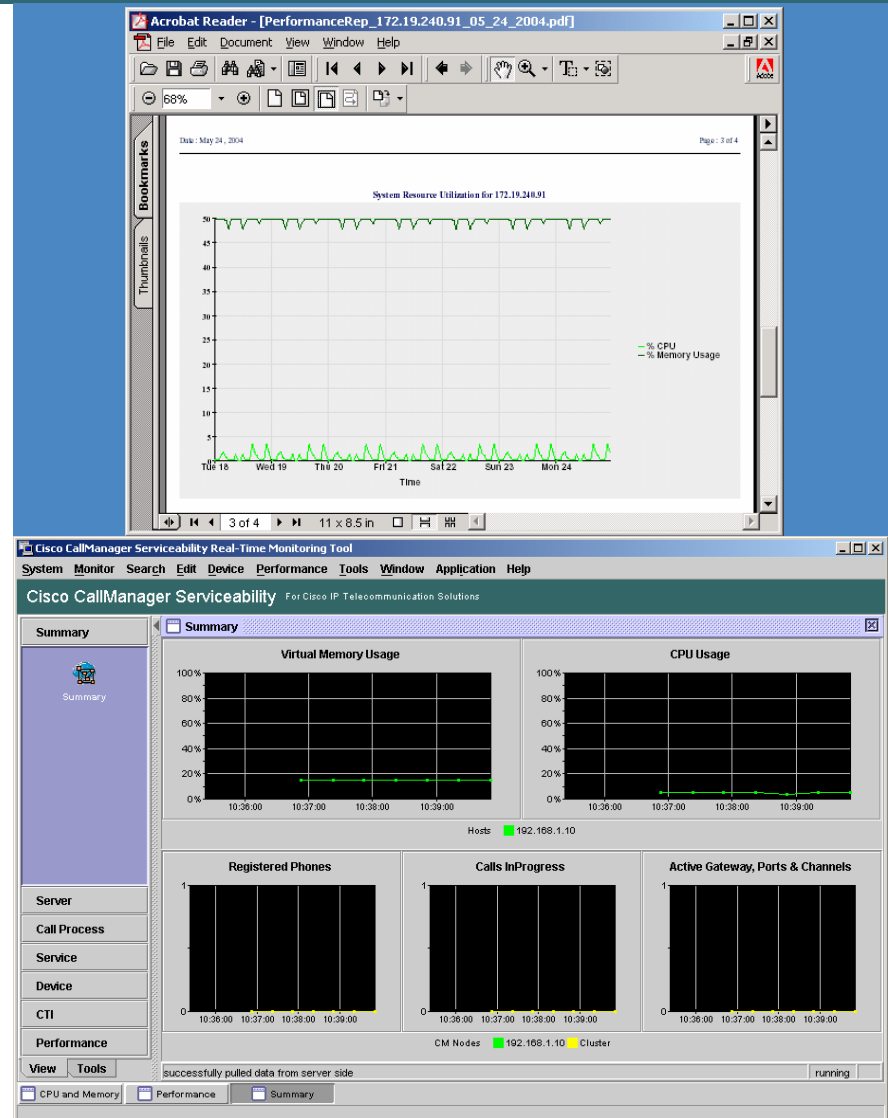
- Provides a real time information about the Cisco Unified CallManager 5.0 system status.
- Provides real time access to Cisco Unified CallManager system traces.
- Java based console available for Linux and Windows systems.

## CiscoWorks IP Communications Operations Manager:

- Provides a unified view of the entire IP Communications infrastructure.
- Presents the current operational status of each element of the IP Communications network.

# Additional Tools to Manage Cisco Unified CallManager 5.0: RTMT

- Win and Linux client
- Integrated TCT, syslog viewer and QRT viewer
- HTTPS support
- OS monitoring
- Cisco Unified CallManager pre-canned DB activity
- SIP phone/protocol activity
- Server based PerfmonCollection (AMC)



# Summary

- **Platform and system management is done via the Cisco Unified CallManager web GUI and network management applications.**
- **Alarms can be stored on different locations.**
- **Traces can be defined on a parameter basis.**
- **The tools menu contains**
- **SNMP Configuration contains tools to manage Cisco Unified CallManager services.**
- **Many CLI commands can be used similar onto the platform administration web GUI.**
- **To manage Cisco Unified CallManager 5.0 are the RTMT and the IP Communications Operations Manager can be used.**

# CISCO SYSTEMS





## **Administering Cisco Unified CallManager 5.0**

# **Backing Up and Restoring Cisco Unified CallManager 5.0**

# Objectives

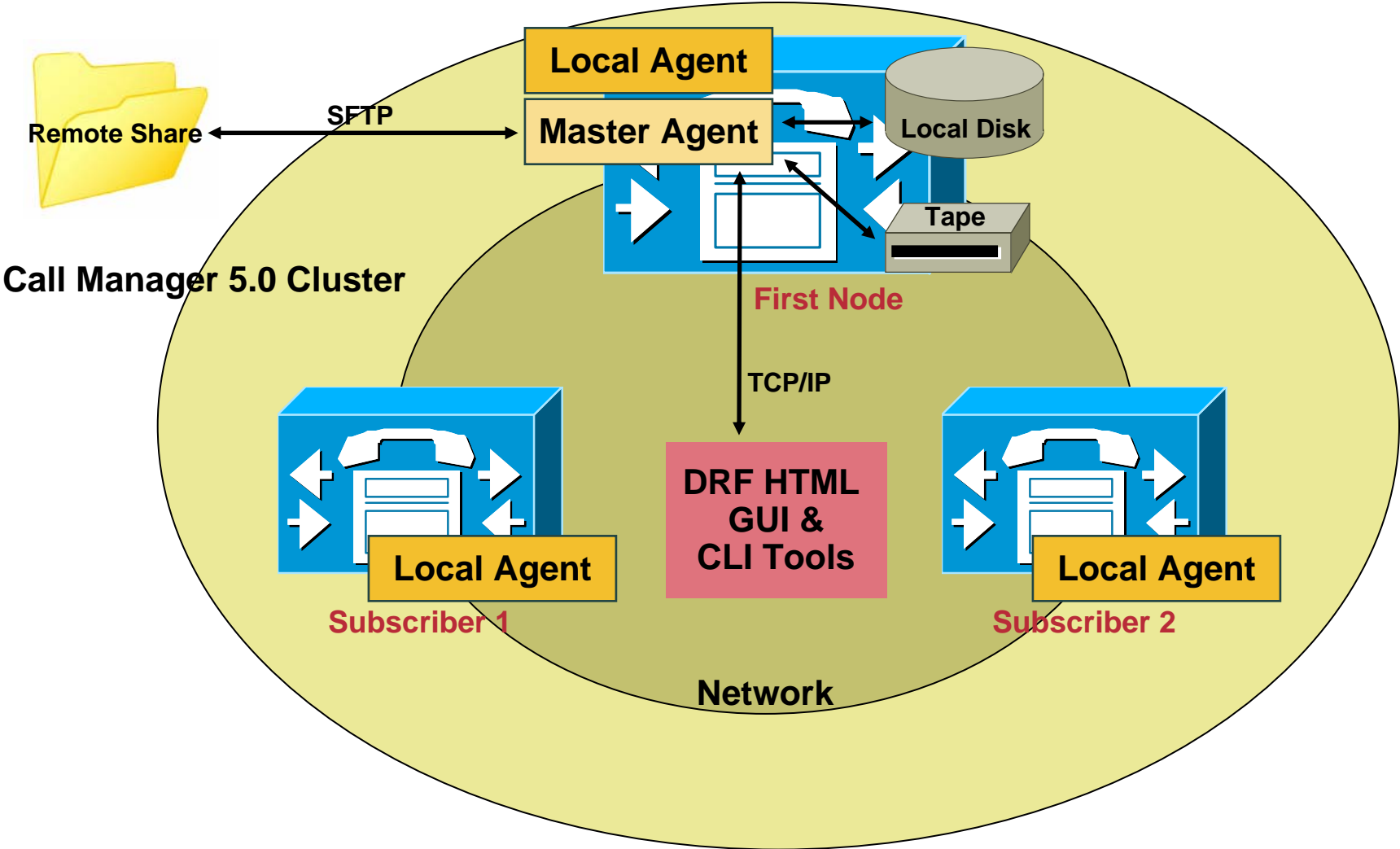
- **Backing Up and Restoring Cisco Unified CallManager 5.0 Overview**
- **CallManager 5.0 Architecture**
- **Backup Menu**
- **Restore Menu**
- **Backup and Restore Procedure**
- **Disaster Recovery**
- **Dual Partitions (Switch Version)**

# Backing Up and Restoring Cisco Unified CallManager 5.0 Overview

- **The Disaster Recovery Framework (DRF) provides backup and restore functionality for Cisco Unified CallManager 5.0.**
- **DRF replaces Backup and Restore System (BARS).**
  - **Like BARS, DRF only backs up the publisher.**
  - **Subscribers will be rebuilt polling their DB from the publisher.**
- **With the dual partition system it is possible to switch between to versions of Cisco Unified CallManager 5.0.**
- **On the web GUI DRF is called Disaster Recovery System (DRS).**



# CallManager 5.0 Architecture



# CallManager 5.0 Architecture

## Disaster Recovery Framework (DRF) backs up features:

- **By selecting a feature all components will be backed up**
  - **Features**
    - **CCM**
    - **CDR\_CAR**
  - **Components**
    - **CCM – CCMDB, PLATFORM, MOH, and others...**
    - **CDRM – CAR, CDR**

# Backup Menu

Disaster Recovery System - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/drf/showhome.do

Navigation Disaster Recovery System Go

Disaster Recovery System For Cisco IP Telecommunication Solutions Logged in as: administrator

Backup Restore Help

- Configure Features
- Storage Location
- Scheduler
- Manual Backup
- History
- Current Status

**Configure backup parameters**

**Start manual backup**

**View backup information**

Administration version: 2.0.0.2-2

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javascript:void(0) 192.168.1.10

# Restore Menu

Disaster Recovery System - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/drf/showhome.do

Navigation Disaster Recovery System Go

Disaster Recovery System For Cisco IP Telecommunication Solutions Logged in as: administrator

Backup Restore Help

Restore Wizard  
History  
Status

**Start system restore**

**View restore information**

System Adminis

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javascript:void(0) 192.168.1.10

# Backup and Restore Procedure

**To backup and restore a Cisco Unified CallManager 5.0 system, the following steps are necessary:**

- 1. Verify that the DRF Master Service is running.**
- 2. Configure backup parameters.**
- 3. Define storage location.**
- 4. Either perform a scheduled or manual backup.**
- 5. Use the restore wizard to restore the system from a backup file.**

# Backup and Restore Procedure: Service Verification

The screenshot shows the Cisco CallManager Serviceability-Control Center interface. It displays several sections of service status:

- Performance and Monitoring:**
  - Cisco CallManager Serviceability RTMT: Running
  - Cisco RTMT Reporter Servlet: Running
  - Cisco Tomcat Stats Servlet: Running
- SOAP Services:**
  - SOAP -Real-Time Service APIs: Running
  - SOAP -Performance Monitoring APIs: Running
  - SOAP -Log Collection APIs: Running
- Backup and Restore Services:**
  - Cisco DRF Master: Running** (highlighted with a red box)
  - Cisco DRF Local: Running
- CDR Services:**
  - Cisco CDR Repository Manager: Running
  - Cisco CDR Agent: Running

At the bottom of the interface, there are buttons for Start, Stop, Restart, and Refresh. The status bar at the bottom right shows the IP address 192.168.1.10.

**Make sure that DRF master service is running on the publisher**

# Backup and Restore Procedure: Feature Backup Selection

Configure Cluster Features - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/drf/configure.do

Navigation Disaster Recovery System Go

Disaster Recovery System For Cisco IP Telecommunication Solutions

Logged in as: administrator

Backup Restore Help

Configure Cluster Features

Status

Status:Ready

Select Features

CCM

CDR\_CAR

Save Select All Clear All

Done 192.168.1.10

Switch to the disaster recovery system ①

Select/deselect feature ②

Save changes ③

# Backup and Restore Procedure: Backup Target Configuration

The screenshot shows the 'Storage Location' configuration page in a Mozilla Firefox browser. The page title is 'Disaster Recovery System For Cisco IP Telecommunication Solutions' and the user is logged in as 'administrator'. The page has a navigation menu with 'Backup', 'Restore', and 'Help' options. The main content area is titled 'Storage Location' and contains the following sections:

- Status:** Status: Ready
- Preserve Tar Files:** Number of backups to store on Network Directory: 2
- Select Destination:**
  - Tape Device: Device name: - Not Selected -
  - Network Directory: Server name: 10.128.128.134, Path name: /upload, User name: ccm50, Password: [masked]

A red box highlights the 'Network Directory' configuration fields. A yellow callout bubble with a circled '1' points to this section with the text 'Configure backup destination'. A 'Save' button is located below the configuration fields, and a yellow callout bubble with a circled '2' points to it with the text 'Save changes'. A status message at the bottom indicates that the settings apply to both Scheduled and Manual Backups.



# Backup and Restore Procedure: Backup Scheduler Configuration

**Scheduler - Mozilla Firefox**  
File Edit View Go Bookmarks Tools Help  
https://192.168.1.10/drf/scheduler.do  
Navigation Disaster Recovery System Go  
Disaster Recovery System For Cisco IP Telecommunication Solutions Logged in as: administrator  
Backup Restore Help

**Scheduler**

Status  
Disabled

Start Backup at

Date 2006 Feb 08  
Time 07 18

Frequency

Once  
 Daily  
 Weekly  
 Monday  Tuesday  Wednesday  Thursday  
 Friday  Saturday  Sunday  
 Monthly

Save Set Default Disable Scheduler **Enable Scheduler**

Done 192.168.1.10

**Define start time and date** ①

**Define backup frequency** ②

**Enable scheduler** ③

# Backup and Restore Procedure: Manual Backup

Manual Backup - Mozilla Firefox

File Edit View Go Bookmarks Tools Help


https://192.168.1.10/drf/backup.do

Navigation Disaster Recovery System Go

Disaster Recovery System For Cisco IP Telecommunication Solutions Logged in as: administrator

Backup Restore Help

**Manual Backup**



**Status**  
Status: Ready

**Features selected**

- ◆ CCM
- ◆ CDR\_CAR

Start Backup

Done 192.168.1.10

**Manually start backup**

# Backup and Restore Procedure: Manual Backup

**Refresh to view actual status**

**Cancel backup**

**Status**

Backup operation in progress for server [CM1], please wait...

**Backup details**

Tar Filename: 2006-02-23-03-40-09.tar  
Storage Location: NETWORK  
Operation: BACKUP  
Percentage Complete: 20%

Feature	Server	Component	Status	Result	Start Time	Log File
CCM	CM1	CDPAGT	100	SUCCESS	Thu Feb 23 03:40:09 PST 2006	<a href="#">2006-02-23-03-40-09_b_cm1_ccm_cdpagt.log</a>
CCM	CM1	CCMPREFS	100	SUCCESS	Thu Feb 23 03:40:10 PST 2006	<a href="#">2006-02-23-03-40-09_b_cm1_ccm_ccmprefs.log</a>
CCM	CM1	CCMDB	0	Active	Thu Feb 23 03:40:10 PST 2006	
CCM	CM1	SYSLOGAGT	0	---	---	
CCM	CM1	BAT	0	---	---	
CCM	CM1	MOH	0	---	---	
CCM	CM1	TFTP	0	---	---	
CCM	CM1	TCT	0	---	---	
CCM	CM1	PLATFORM	0	---	---	
CDR_CAR	CM1	CDRREP	0	---	---	

Done 192.168.1.10

# Backup and Restore Procedure: Restore Wizard

Step1 Restore - Choose Storage Location - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/DRF/restore.do

Navigation Disaster Recovery System Go

Disaster Recovery System For Cisco IP Telecommunication Solutions Logged in as: administrator

Backup Restore Help

Step1 Restore - Choose Storage Location

Status

Status:Ready

Tape Device

Device name - Not Selected -

Network Directory

Server name 10.128.128.134

Path name /upload

User name ccm50bc

Password \*\*\*\*\*

Next Cancel

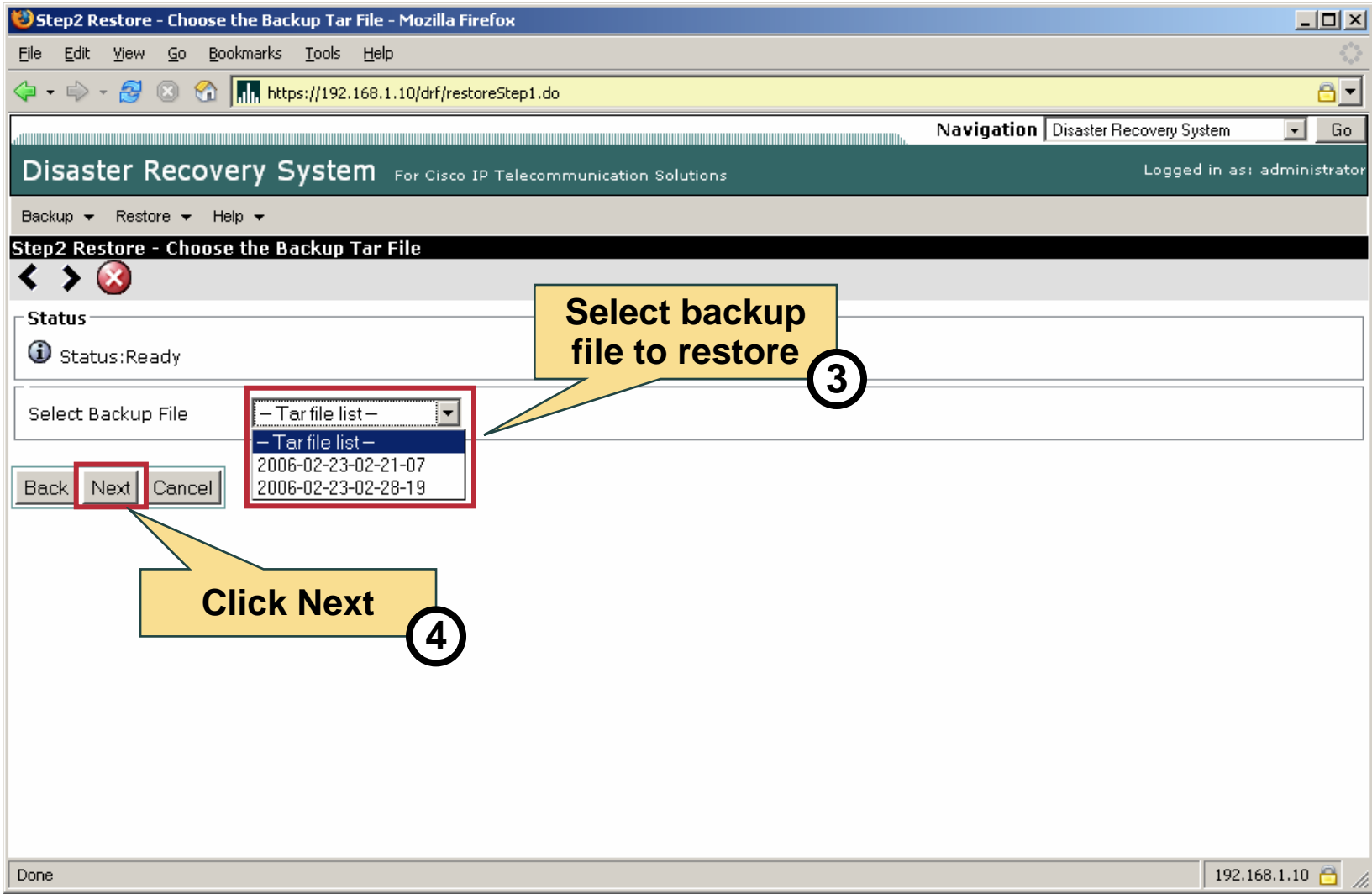
Done

192.168.1.10

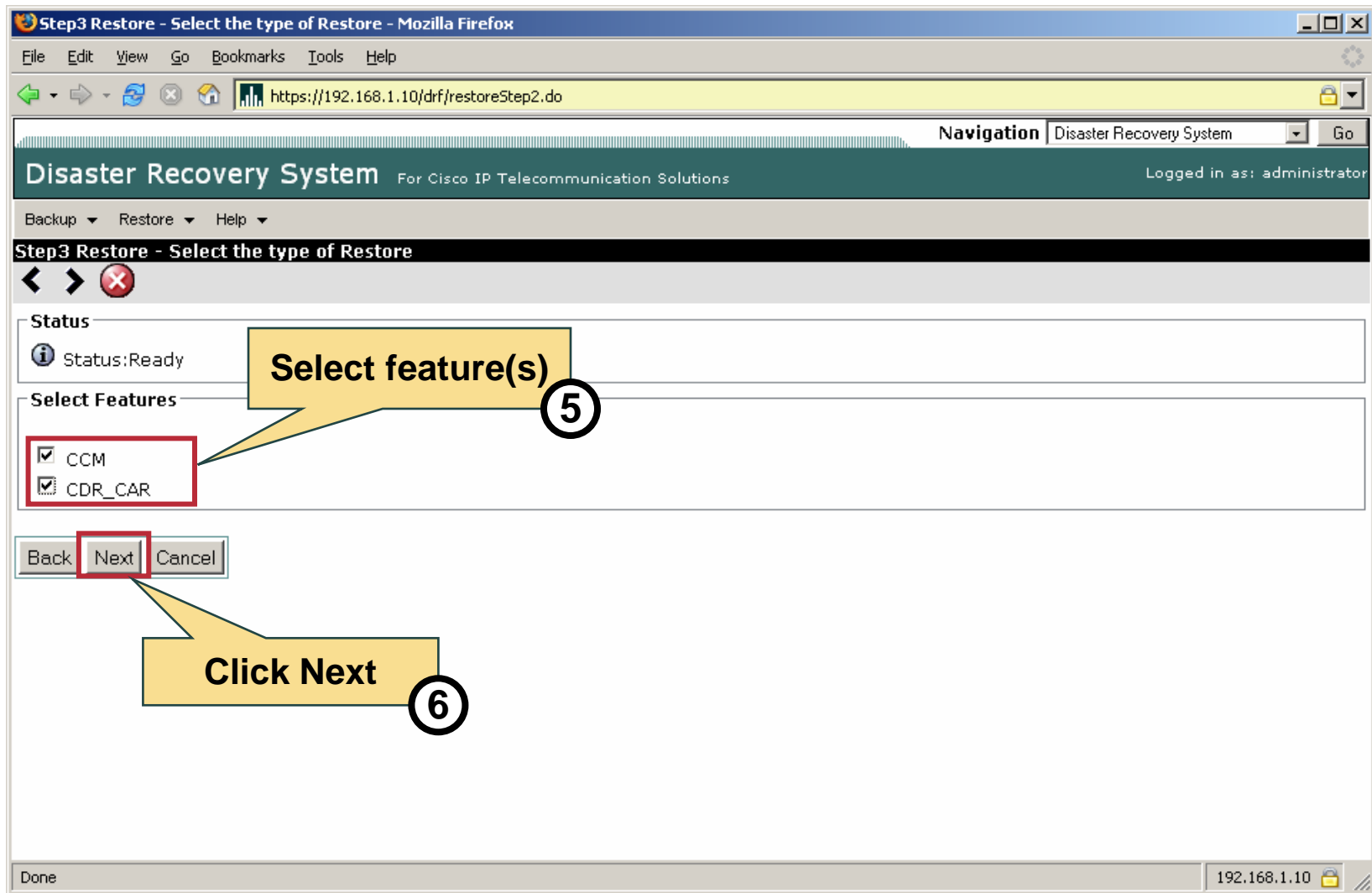
Select backup file source to restore ①

Click Next ②

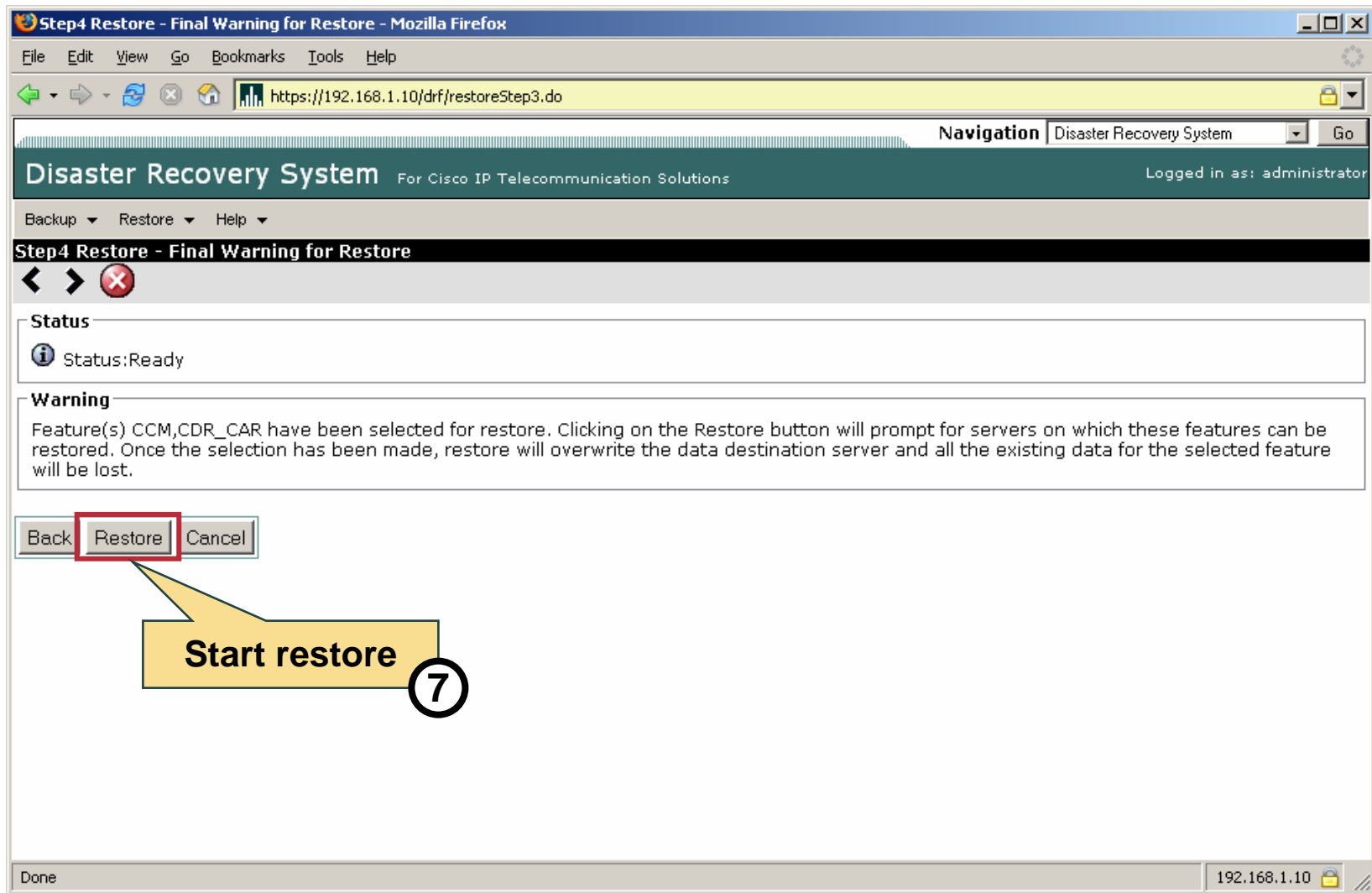
# Backup and Restore Procedure: Restore Wizard



# Backup and Restore Procedure: Restore Wizard



# Backup and Restore Procedure: Restore Wizard



# Backup and Restore Procedure: Restore Wizard

Restore Status - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/DRF/restorestatus.do

Navigation Disaster Recovery System Go

Disaster Recovery System For Cisco IP Telecommunication Solutions Logged in as: administrator

Backup Restore

**Refresh to view actual status**

**8**

**Restore Status**

Status

Restoring server [CM1], please wait...

Restore details

Tar Filename: 2006-02-23-02-21-07.tar  
Storage Location: NETWORK  
Operation: RESTORE  
Percentage Complete: 25%

Feature	Server	Component	Status	Result	Start Time	Log File
CCM	CM1	CDPAGT	100	SUCCESS	Thu Feb 23 03:16:37 PST 2006	<a href="#">2006-02-23-02-21-07 r cm1 ccm cdpagt.log</a>
CCM	CM1	CCMPREFS	100	SUCCESS	Thu Feb 23 03:18:04 PST 2006	<a href="#">2006-02-23-02-21-07 r cm1 ccm ccmprefs.log</a>
CCM	CM1	CCMDB	50	Active	Thu Feb 23 03:18:04 PST 2006	
CCM	CM1	BAT	0	---	---	
CCM	CM1	MOH	0	---	---	
CCM	CM1	TFTP	0	---	---	
CCM	CM1	TCT	0	---	---	
CCM	CM1	PLATFORM	0	---	---	
CCM	CM1	SYSLOGAGT	0	---	---	
CDR_CAR	CM1	CDRREP	0	---	---	

Click on the filename to view the log file

Done 192.168.1.10



# Disaster Recovery

The following disaster recovery commands are available on the Cisco Unified CallManager 5.0 CLI:

```
admin:utils disaster_recovery ?
      utils disaster_recovery status
      utils disaster_recovery cancel_backup
      utils disaster_recovery show_tapeid
      utils disaster_recovery show_registration
      utils disaster_recovery show_backupfiles*
      utils disaster_recovery backup*
      utils disaster_recovery restore*

admin:utils disaster_recovery
```

# Disaster Recovery

admin:

```
utils disaster_recovery show_tapeid
```

- **This command will display a list of available tape devices on the publisher running the Master Agent.**
- **This device name must be used when backing up or restoring from tape.**
- **The format of the name will be:**
  - **/dev/nst0**

# Disaster Recovery

admin:

```
utils disaster_recovery backup tape <tapeid>  
utils disaster_recovery backup <path> <sftp server> <userid>
```

- **The CLI backup command backs up all features.**
- **This command will initiate a backup to tape or network sftp server.**
- **The <tapeid> parameter is obtained from the show\_tapeid command.**
- **When backing up to the network the following information is required:**
  - **Network path**
  - **UserID to access network path**

# Disaster Recovery

admin:

```
disaster_recovery cancel_backup Y
```

- **This command will cancel an active backup after the current component completes its' backup.**

admin:

```
utils disaster_recovery status Restore  
utils disaster_recovery status Backup
```

- **This command will display current backup or restore status.**

# Disaster Recovery

admin:

```
utils disaster_recovery show_backupfiles tape <tapeid>  
utils disaster_recovery show_backupfiles <path> <sftp server> <userid>
```

- This command is used to list all backup sets on tape, local disk, or network SFTP server.
- The output of this command is required by the “restore” command.

# Disaster Recovery

admin:

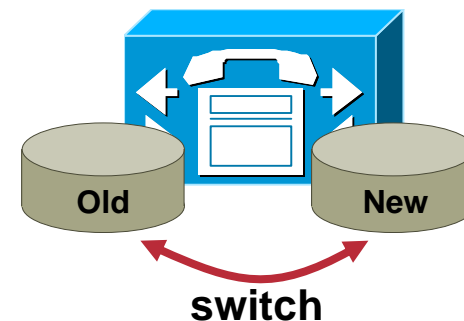
```
utils disaster_recovery restore tape server tarfile <tapeid>  
utils disaster_recovery restore network <server> <tarfilename>  
<path> <sftp server> <userid>
```

- This command is used to restore all features to a specific server.
- The <tapeid> parameter can be obtained from the “show\_tapeid” command.
- The first <server> parameter indicates which server should be restored.

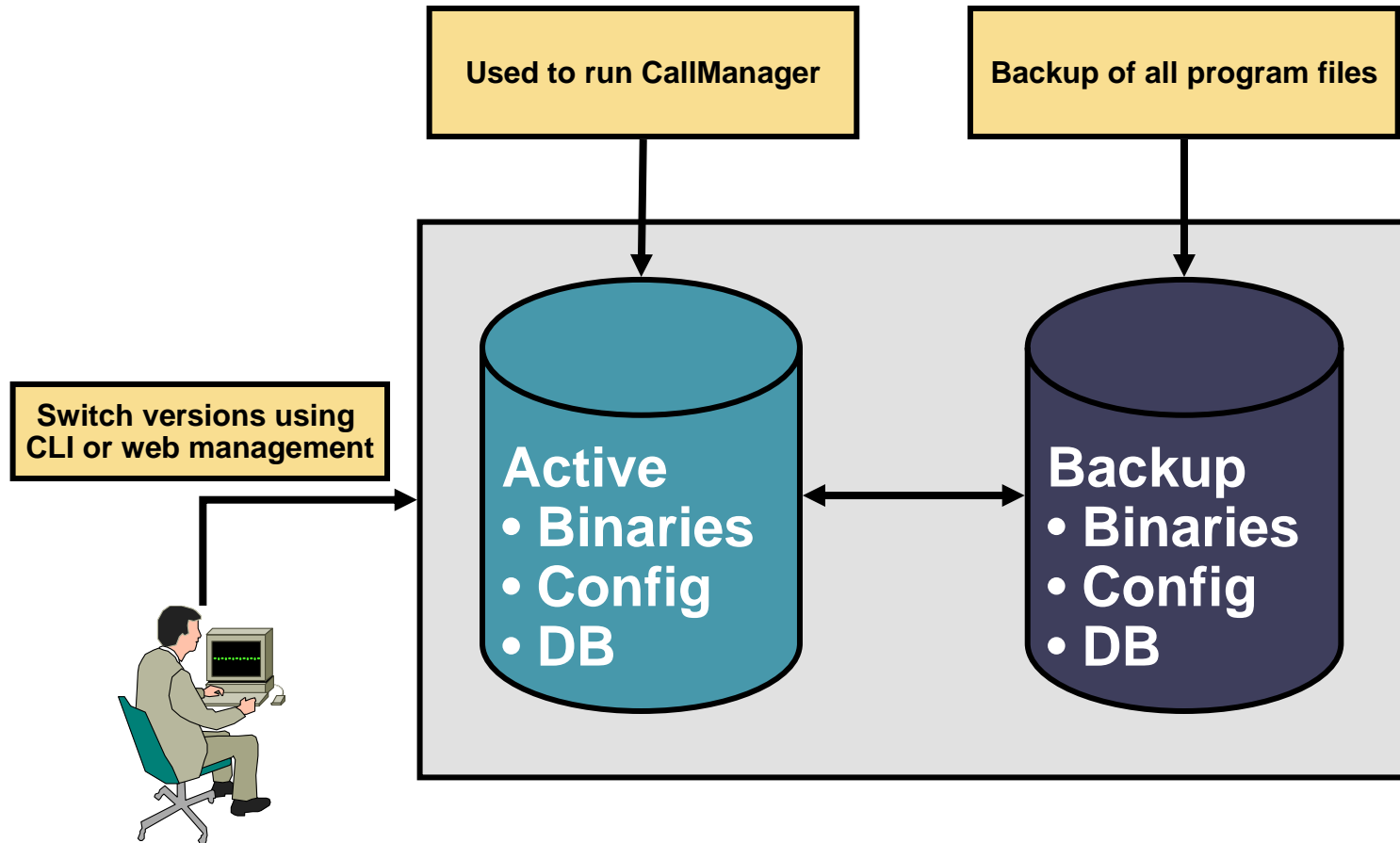
# Dual Partitions and Switch Version

## Cisco Unified CallManager supports dual partitions to switch versions:

- When upgrading to a new version, the actual version is stored on the second partition.
- The system can be switches between those two partitions.
- It is necessary to reload the system to switch to the other partition.



# Dual Partitions (Switch Version)





# Summary

- **Cisco Unified CallManager 5.0 uses DRF and dual partitions for recovery purposes.**
- **The master agent runs on the publisher and performs the actual backup.**
- **The backup menu contains all necessary parts to backup the system using DRF.**
- **The restore menu contains all necessary parts to restore the system using DRF.**
- **To backup and restore a Cisco Unified CallManager 5.0 system backup files are used.**
- **The disaster recovery allows to backup a system using the Cisco Unified CallManager 5.0 CLI.**
- **Dual partitions allow to switch between two Cisco Unified CallManager versions.**

# CISCO SYSTEMS





# Deploying Cisco Unified CallManager 5.0 Endpoints

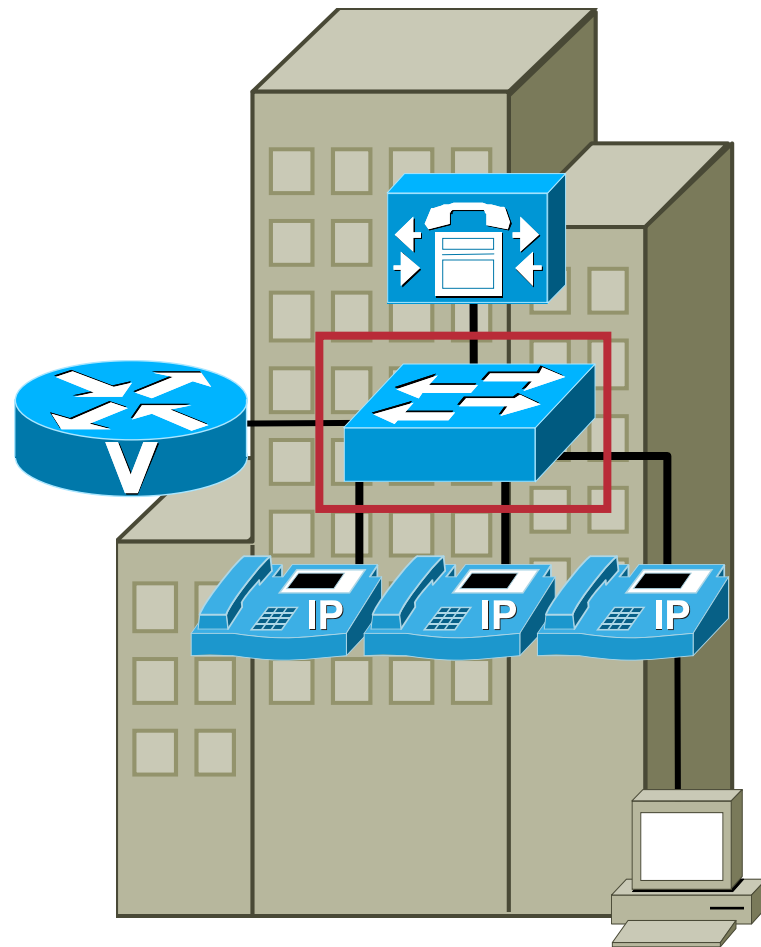
## Configuring Catalyst Switches for Endpoints

# Objectives

- **Catalyst Switch Role in IP Telephony**
- **Powering the Cisco IP Phone**
- **Types of PoE Delivery**
- **Catalyst Family of PoE Switches**
- **Configuring PoE**
- **Configuring Dual VLANs**
- **Configuring Class of Service**

# Catalyst Switch Role in IP Telephony

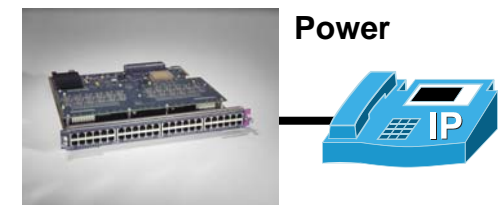
- Supplies inline power to IP Phones
- Supports voice and data VLANs on a single port
- Prioritizes voice traffic with Class of Service (CoS) marking



# Powering the Cisco IP Phone

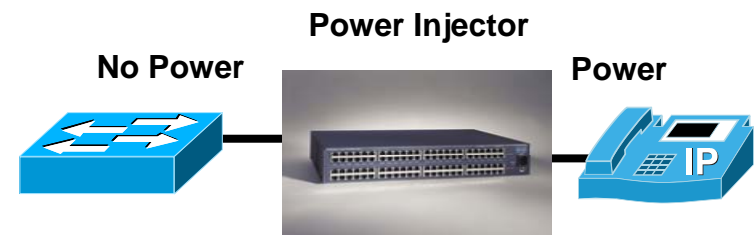
## Power over Ethernet (PoE):

- Need PoE line cards or PoE ports for Cisco Catalyst switches
- Delivers -48 V DC over data pairs (pins 1, 2, 3, and 6)



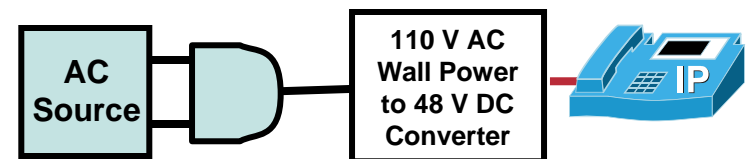
## Midspan power injection:

- Needs external power source equipment
- Delivers -48 V DC over spare pairs (pins 4, 5, 7, and 8)



## Wall power:

- Needs DC converter for connecting IP Phone to wall outlet



# Types of PoE Delivery

## **Cisco original implementation:**

- **Provides -48 V DC at up to 6.3 to 7.7 W per port over data pins 1, 2, 3, and 6.**
- **Supports most Cisco devices (IP Phones and wireless access points).**
- **Uses a Cisco proprietary method to determine if an attached device requires power. Power is delivered only to devices that require power.**

# Types of PoE Delivery

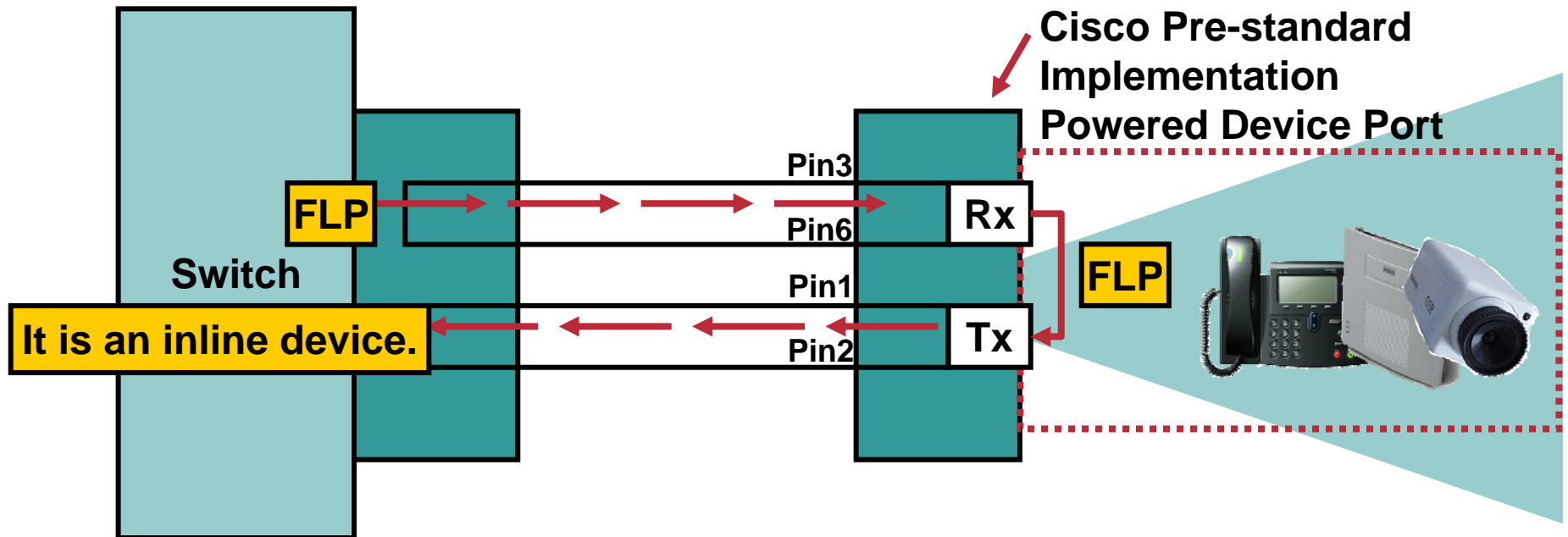
## **IEEE 802.3af Power over Ethernet:**

- **Specifies -48 V DC at up to 15.4 W per port over data pins 1, 2, 3, and 6 or spare pins 4, 5, 7, and 8.**
- **Enables a new range of Ethernet-powered devices because of increased power.**
- **Standardizes the method to determine if an attached device requires power. Power is delivered only to devices that require power.**
- **Has several optional elements, including power classification.**
- **Caution: Verify that the deployed switch actually can supply enough power to the deployed number and types of IP phones.**



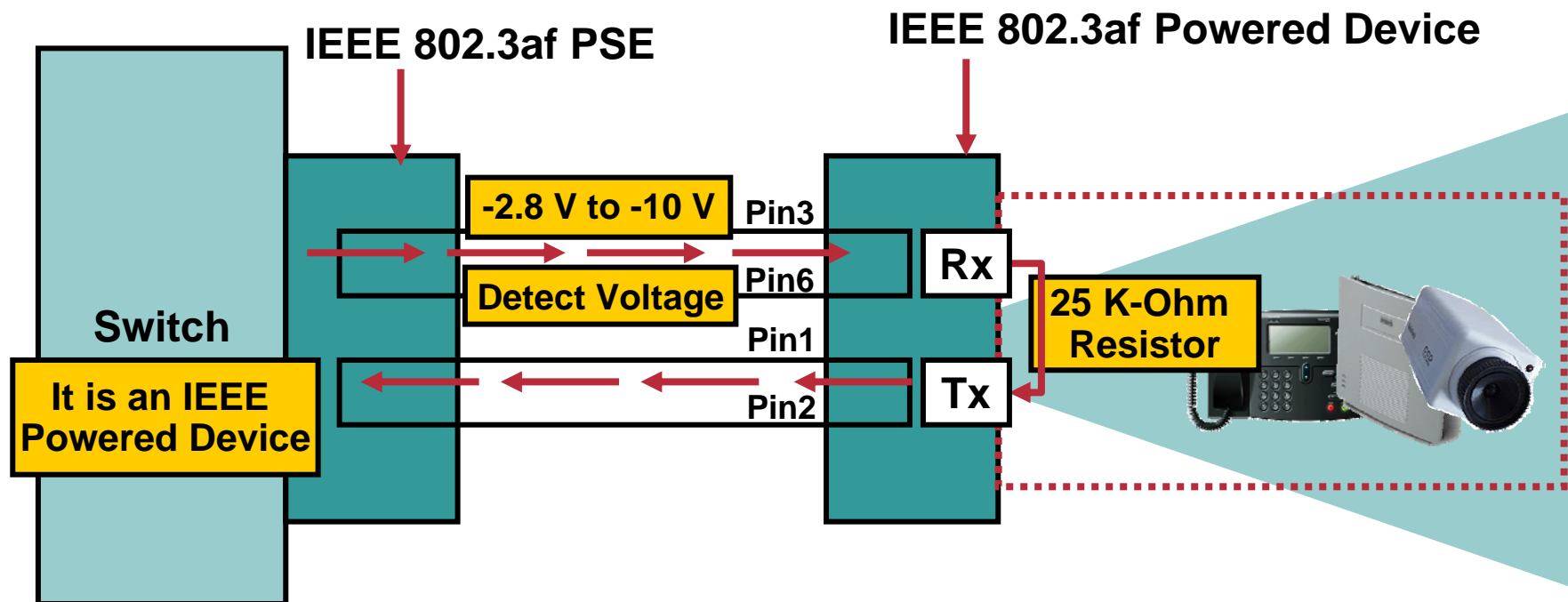
# Types of PoE Delivery

## Cisco pre-standard device detection:



# Types of PoE Delivery

## IEEE 802.3af device detection:



# Configuring PoE

CatOS>(enable)

```
set port inlinepower <mod/port> ?  
  auto   Port inline power auto mode  
  off    Port inline power off mode
```

- **Enable/disable PoE on Cisco CatOS switches**

IOS(config-if)#

```
power inline ?  
  auto   Automatically detect and power inline devices  
  delay  Inline power delay timer setting  
  never  Never apply inline power
```

- **Enable/disable PoE on native Cisco IOS switches**

# Configuring PoE

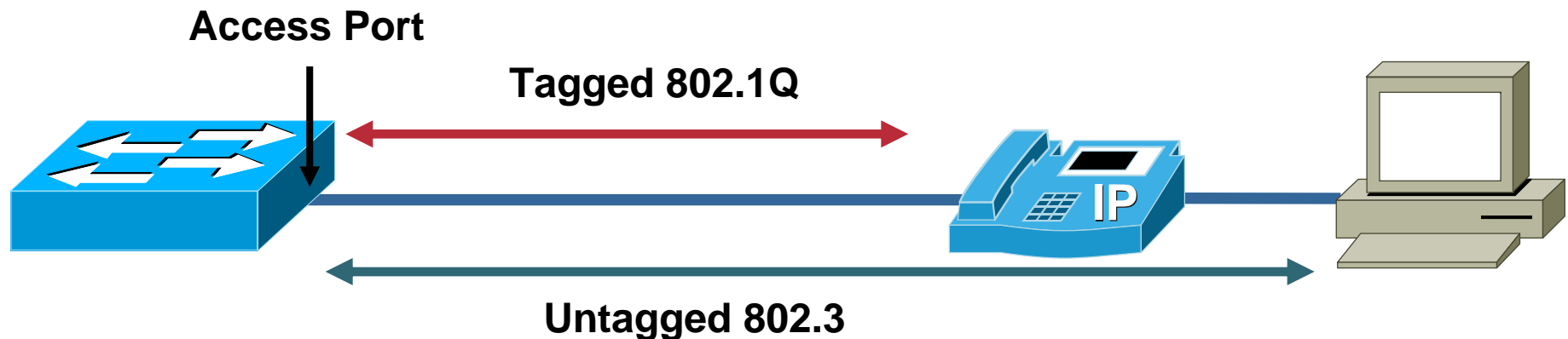
```
CatOS>(enable) show port inline power 7
Default Inline Power allocation per port: 10.000 Watts (0.23 Amps @42V)
Total inline power drawn by module 7: 75.60 Watts (1.80 Amps @42V)
Port      InlinePowered      PowerAllocated
         Admin      Oper      Detected      mWatt      mA @42V
-----
7/1      auto      off      no      0      0
7/2      auto      on       yes      6300     150
7/3      auto      on       yes      6300     150
7/4      auto      off      no      0      0
7/5      auto      off      no      0      0
7/6      auto      off      no      0      0
7/7      auto      off      no      0      0
```

```
IOS> show power inline
Interface      Admin      Oper      Power ( mWatt )      Device
-----
FastEthernet9/1  auto      on       6300      Cisco 6500 IP Phone
FastEthernet9/2  auto      on       6300      Cisco 6500 IP Phone
FastEthernet9/3  auto      off      0         n/a
```

# Configuring Dual VLANs

## A Multi-VLAN Access Port:

- Access ports that are able to handle two types of VLANs.
- Configured with an access (data) VLAN and a voice (auxiliary) VLAN.



# Configuring Dual VLANs

## Configuring voice VLANs on native Cisco IOS using Catalyst 3560, 3750, Catalyst 4500, Catalyst 6500:

```
IOS(config)#interface FastEthernet0/1
IOS(config-if)#switchport mode access
IOS(config-if)#switchport voice vlan 261
IOS(config-if)#switchport access vlan 262
IOS(config-if)#spanning-tree portfast
IOS(config-if)#
IOS(config)#show interfaces fa0/4 switchport
Name: Fa0/4
Switchport: Enabled
Administrative Mode: static access
Operational Mode: static access
Administrative Trunking Encapsulation: negotiate
Operational Trunking Encapsulation: native
Negotiation of Trunking: Off
Access Mode VLAN: 262 (VLAN0262)
Trunking Native Mode VLAN: 1 (default)
Voice VLAN: 261 (VLAN0261)
```

# Configuring Dual VLANs

## Configuring voice VLANs on Cisco CatOS:

```
Console>(enable) set port auxiliaryvlan 2/1-3 222
Auxiliaryvlan 222 configuration successful.
AuxiliaryVlan AuxVlanStatus Mod/Ports
```

```
-----
222          active          1/2,2/1-3
```

```
Console> (enable)
```

```
Console> (enable) show port auxiliaryvlan 222
```

```
AuxiliaryVlan AuxVlanStatus Mod/Ports
```

```
-----
222          active          1/2,2/1-3
```

```
Console> (enable)
```

```
Console> (enable) show port 2/1
```

```
...
```

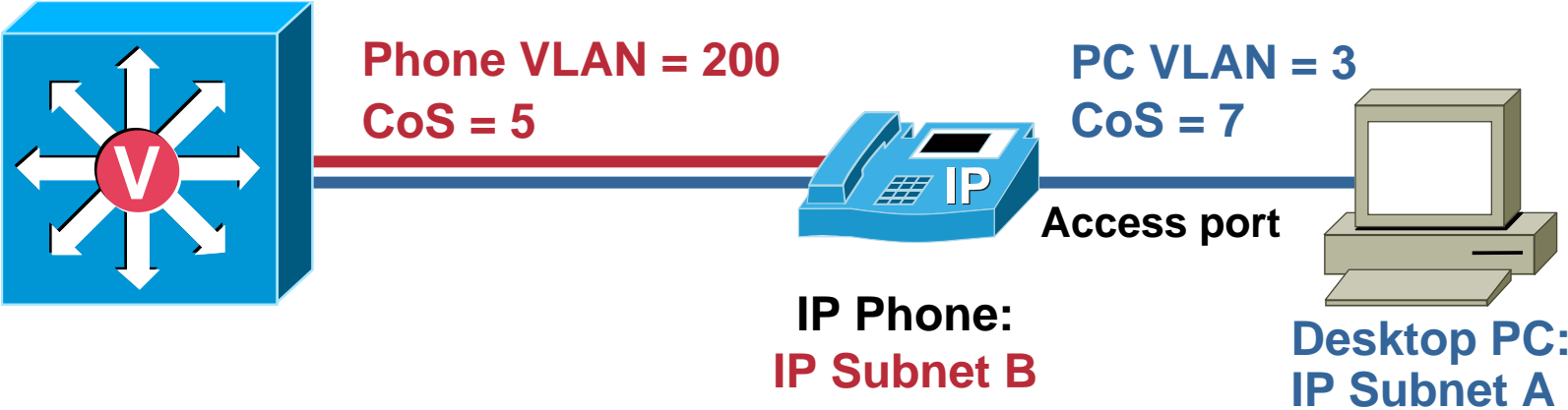
```
Port  AuxiliaryVlan  AuxVlan-Status
```

```
-----
2/1  222          active
```

```
...
```

# Configuring Class of Service

## Extension of the trusted boundary to the IP Phone:



PC is not trusted  
CoS set to 0 (normal)



PC is not trusted  
CoS set to 2



PC is trusted





# Configuring Class of Service

CatOS>(enable)

```
set port qos mod/port cos cos-value
```

- **Set the default value for all packets that have arrived through an untrusted port.**

CatOS>(enable)

```
set port qos mod/ports... trust-ext {trusted | untrusted}
```

- **Allows you to trust or not trust (set to 0) the CoS assigned to the device attached to the IP Phone.**

IOS(config-if)#

```
switchport priority extend {cos value | trust}
```

- **Choose to modify, ignore, or trust the CoS of the attached device.**

# Summary

- **Cisco voice-capable switches support three primary feature sets that can assist with an IP telephony deployment: POE, dual VLANs, and CoS.**
- **Most Cisco IP Phone models are capable of using three different options for power.**
- **Two types of inline power delivery are the Cisco pre-standard implementation and IEEE 802.3af PoE.**
- **The Cisco Catalyst 6500, 4500, 3750, and 3560 switches support 802.3af, Cisco PoE, dual VLANs, and CoS.**
- **Inline power can be enabled or disabled using IOS or CatOS commands.**
- **Using dual VLANs on a single-port Cisco Catalyst switch improves network scalability.**
- **QoS can be used to mark traffic on voice/data ports individually.**

# CISCO SYSTEMS





## Deploying Cisco Unified CallManager 5.0 Endpoints

# Configuring SCCP Endpoints in Cisco Unified CallManager 5.0

# Objectives

- **Endpoint Configuration Overview**
- **Configuring Device Pools**
- **IP Phone Button Templates**
- **Manual IP Phone Configuration**
- **Line Configuration**
- **Configuring IP Phone Auto-Registration**
- **Verify Endpoint Configuration**

# Endpoint Configuration Overview

**SCCP Phones on Cisco Unified CallManager 5.0 are configured by using the following functionality:**

- **Cisco Unified CallManager device pools allow configuration of global settings for all devices within that pool.**
- **IP phone button templates define the function of phone buttons.**
- **IP phones and IP phone lines can be configured manually.**
- **Auto-registration allows automatic phone registration and configuration.**

# Configuring Device Pools

**These steps are necessary to configure device pools:**

- **Add a new device pool.**
- **Enter a name for the device pool.**
- **Configure Cisco Unified CallManager values for the device pool.**
- **Select a default or individually configured softkey template that is used by phones within that device pool.**
- **Define SRST reference**
  - **Select SRST reference should be used by phones within that device pool.**
  - **If the phones should not use SRST select “disable”.**
- **Save changes and reset the device pool.**

# Configuring Device Pools

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10/ccmadmin/devicePoolFindList.do`. The page title is "Find and List Device Pools". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The main content area is titled "Find and List Device Pools" and contains a plus sign icon. Below this, the "Status" section shows "0 records found". The "Search Options" section includes a search bar with "Device Pool Name" selected, "begins with" as the filter, and a "Find" button. A checkbox for "Search Within Results" is also present. The "Search Results" section displays the message: "No active query. Please enter your search criteria using the options above." A red box highlights the "Add New" button, and a yellow callout bubble with the text "Add a new device pool" and a circled "1" points to it.



# Configuring Device Pools

The screenshot shows the 'Device Pool Configuration' page in a web browser. The page title is 'Device Pool Configuration' and the status is 'Ready'. The device pool name is 'SanJose' with 2 members. The 'Device Pool Settings' section includes several dropdown menus: 'Device Pool Name\*' (SanJose), 'Cisco CallManager Group\*' (Default), 'Date/Time Group\*' (CMLocal), 'Region\*' (Default), 'Softkey Template\*' (Standard User), and 'SRST Reference\*' (Use Default Gateway). The 'Multilevel Precedence and Preemption (MLPP) Information' section is partially visible at the bottom.

Callout 1: Save changes

Callout 2: Enter device pool name

Callout 3: Configure CallManager values for device pool

Callout 4: Select softkey template for phones in device pool.

Callout 5: Select SRST GW for phones in device pool.

Callout 6: Save changes

Callout 7: Reset Pool

# IP Phone Button Templates

- **Default 7960/7961 template is 2 lines, 4 speed dials.**
- **Buttons can be configured for**
  - **Line**
  - **Speed dial**
  - **Service URL**
  - **Privacy on/off**
  - **Speed dial Busy Lamp Field (BLF)**
- **The first phone button needs to be a line button.**
- **All other phone buttons can be configured individually for any possible function.**



# IP Phone Button Templates

## Handling IP phone button template names:

- **Template updates affect any IP Phone that uses that template.**
- **Renaming a template does not affect the IP Phones using that template.**
- **Templates assigned to one or more devices can not be deleted.**



# IP Phone Button Templates

**These tasks are necessary to add a new personalized phone button template:**

- **Search for any existing phone button templates.**
- **Find the template that most closely matches what you want to do and copy it.**
- **Enter a name for the new template.**
- **Change the buttons to meet the requirements.**
- **Save changes.**

# IP Phone Button Templates

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10/ccmadmin/phoneTemplateFindList.do`. The page title is "Find and List Phone Button Templates". The interface includes a navigation menu with options like "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The main content area is titled "Find and List Phone Button Templates" and features a search form. The search form has a "Find" button highlighted with a red box. A yellow callout bubble with a circled "1" points to the "Find" button, containing the text "Search for existing templates". The search options include a dropdown menu set to "begins with", a text input field, and a "Search Within Results" checkbox. The search results section shows "No active query. Please enter your search criteria using the options above." and an "Add New" button. The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# IP Phone Button Templates

Find and List Phone Button Templates - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/phoneTemplateFindList.do?%3C%3DreqParams%3E&recCnt=0&colCnt=3

<a href="#">Standard 7910</a>	SCCP	
<a href="#">Standard 7911</a>	SCCP	
<a href="#">Standard 7911 SIP</a>	SIP	
<a href="#">Standard 7912 SCCP</a>	SCCP	
<a href="#">Standard 7912 SIP</a>	SIP	
<a href="#">Standard 7920</a>	SCCP	
<a href="#">Standard 7935</a>	SCCP	
<a href="#">Standard 7936</a>	SCCP	
<a href="#">Standard 7940 SCCP</a>	SCCP	
<a href="#">Standard 7940 SIP</a>	SIP	
<a href="#">Standard 7941 SCCP</a>	SCCP	
<a href="#">Standard 7941 SIP</a>	SIP	
<a href="#">Standard 7941G-GE SCCP</a>	SCCP	
<a href="#">Standard 7941G-GE SIP</a>	SIP	
<a href="#">Standard 7960 SCCP</a>	SCCP	
<a href="#">Standard 7960 SIP</a>	SIP	
<a href="#">Standard 7961 SCCP</a>	SCCP	
<a href="#">Standard 7961 SIP</a>	SIP	
<a href="#">Standard 7961G-GE SCCP</a>	SCCP	
<a href="#">Standard 7961G-GE SIP</a>	SIP	
<a href="#">Standard 7970 SCCP</a>	SCCP	
<a href="#">Standard 7970 SIP</a>	SIP	
<a href="#">Standard 7971 SCCP</a>	SCCP	
<a href="#">Standard 7971 SIP</a>	SIP	
<a href="#">Standard 7985</a>	SCCP	
<a href="#">Standard ATA 186</a>	SCCP	
<a href="#">Standard Analog</a>	SCCP	
<a href="#">Standard IP-STE</a>	SCCP	
<a href="#">Standard VGC Phone</a>	SCCP	

Done 192.168.1.10

Copy existing templates 2

# IP Phone Button Templates

Phone Button Template Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/phoneTemplateEdit.do?clone=1&key=5b45f538-1b5f-4b13-9419-8362d202c14d

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Phone Button Template Configuration Related Links: Back To Find/List Go

Status

Status: Ready

Phone Button Template Information

Button Template Name \* 7960-1line-5blf SCCP

Save

\*- indicates required

\*\* indicates that this field is configurable

Enter name for new template 3

Click save to create new template 4

Done 192.168.1.10

# IP Phone Button Templates

**Save changes** (6)

**Select button feature** (5)

Phone Button Template Configuration - Mozilla Firefox  
https://192.168.1.10/ccmadmin/phoneTemplateSave.do

Navigation Cisco CallManager Administration Go

CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

em Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Phone Button Template Configuration Related Links: Back To Find/List Go

Status  
Add successful

Phone Button Template Information  
Button Template Name \* 7960-1line-5blf SCCP

Button Information

Button	Feature	Label
1	Line **	
2	Speed Dial BLF	
3	Speed Dial BLF	Speed Dial BLF
4	Speed Dial BLF	Speed Dial BLF
5	Speed Dial BLF	Speed Dial BLF
6	Speed Dial	Speed Dial 4
7	Speed Dial	None
8	Line	None
9	Privacy	None
	Service URL	None
	Speed Dial BLF	
Done	None	



# Manual IP Phone Configuration

**These are the minimum steps required to manually add an IP phone on Cisco Unified CallManager 5.0:**

- **Add a new phone.**
- **Select the phone type.**
- **Select the phone protocol.**
- **Perform the basic phone configuration.**

# Manual IP Phone Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10/ccmadmin/phoneFindList.do`. The page title is "Find and List Phones - Mozilla Firefox". The navigation menu includes "Cisco CallManager Administration" and "Go". The user is logged in as "CCMAdministrator". The main content area is titled "Find and List Phones" and features a search form with the following elements:

- Search Options:** A form with "Find Phone where" set to "Device Name", "begins with" set to an empty field, a "Find" button, and a "Search Within Results" checkbox.
- Search Results:** A message stating "No active query. Please enter your search criteria using the options above." Below this message is an "Add New" button, which is highlighted with a red box and a yellow callout box containing the text "Add a new phone" and a circled "1".

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# Manual IP Phone Configuration

**Add a New Phone - Mozilla Firefox**  
File Edit View Go Bookmarks Tools Help  
https://192.168.1.10/ccmadmin/phoneEdit.do

**Cisco CallManager Administration** For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator  
System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

**Add a New Phone** Related Links: Back To Find/List Go

Status  
Select the type of phone you would like to create  
Phone Type\*  
Next  
\*- indicates  
- Device re

Done 192.168.1.10

**Click next** ③

**Select phone type** ②

# Manual IP Phone Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows `https://192.168.1.10/ccmadmin/phoneEdit.do`. The page title is "Cisco CallManager Administration" and it shows the user is logged in as "CCMAdministrator".

The "Phone Configuration" section is active, with a "Next" button highlighted in a red box. A yellow callout bubble with the text "Click next" and a circled number "5" points to the "Next" button. Another yellow callout bubble with the text "Select protocol" and a circled number "4" points to a dropdown menu for "Select the device protocol". The dropdown menu is open, showing three options: "SCCP", "SCCP", and "SIP".

Below the form, there are two informational messages:

- \*** - indicates required item.
- Device reset is not required for changes to Packet Capture Mode and Packet Capture Duration.

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# Manual IP Phone Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10/ccmadmin/phoneEdit.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The "Phone Configuration" section is active, showing the following configuration details:

- Status:** Ready
- Phone Type:** Product Type: Cisco 7960, Device Protocol: SCCP
- Device Information:**
  - MAC Address\*: 0017ED2337A1
  - Description: SEP0017ED2337A1
  - Device Pool\*: SanJose
  - Phone Button Template\*: 7960-1line-4blf-1service SCCP
  - Softkey Template: < None >
  - Common Phone Profile\*: Standard Common Phone Profile
  - Calling Search Space: < None >
  - AAR Calling Search Space: < None >
  - Media Resource Group List: < None >
  - User Hold Audio Source: < None >

Three callout boxes with numbered circles provide instructions:

- Box 6: "Enter MAC of phone" points to the MAC Address field.
- Box 7: "Select device pool" points to the Device Pool dropdown menu.
- Box 8: "Select template" points to the Phone Button Template dropdown menu.

# Manual IP Phone Configuration

The screenshot shows the 'Phone Configuration - Mozilla Firefox' window. The address bar displays 'https://192.168.1.10/ccmadmin/phoneEdit.do'. The configuration form includes several dropdown menus and checkboxes. The 'SCCP Phone Security Profile\*' dropdown is highlighted with a red box, and a yellow callout bubble with the text 'Select security profile' and a circled '9' points to it. The selected option in the dropdown is 'Non Secure SCCP Profile auth by String'. Other visible options include '< None >', 'None', '0', 'Standard Presence group', and '< None >'. Checkboxes for 'Retry Video Call as Audio', 'Ignore Presentation Indicators (internal calls only)', 'Allow Control of Device from CTI', 'Unattended Port', 'Require DTMF Reception', and 'RFC2833 Disabled' are also present.

User Hold Audio Source	< None >
Network Hold Audio Source	< None >
Location*	Hub_None
User Locale	< None >
Network Locale	< None >
Built In Bridge*	Default
Privacy*	Default
Owner User ID	< None >
Phone Load Name	
<input checked="" type="checkbox"/> Retry Video Call as Audio	
<input type="checkbox"/> Ignore Presentation Indicators (internal calls only)	
<input checked="" type="checkbox"/> Allow Control of Device from CTI	
<b>Protocol Specific Information</b>	
Packet Capture Mode*	None
Packet Capture Duration	0
Presence Group*	Standard Presence group
SCCP Phone Security Profile*	Non Secure SCCP Profile auth by String
SUBSCRIBE Calling Search Space	< None >
<input type="checkbox"/> Unattended Port	
<input type="checkbox"/> Require DTMF Reception	
<input type="checkbox"/> RFC2833 Disabled	
<b>Expansion Module Information</b>	

# Manual IP Phone Configuration

## Phone settings sections:

Section	Function
Device Information	Device parameters.
Protocol Specific Information	SCCP/SIP related settings.
Expansion Module Information	7914 expansion module configuration.
External Data Locations	URLs for external data sources.
Extension Information	Extension mobility configuration.
Certification Authority Proxy Function	CAPF phone configuration.
MLPP Information	Multilevel Precedence and Preemption phone configuration.
Secure Shell	Secure shell user information.
Product Specific Configuration	Device configuration specific for the phone type.

# Line Configuration

**You must configure at least one line on the phone to complete the phone configuration**

- **Add a new directory number (DN) for the phone.**
- **Enter the DN.**
- **Enter a description for the DN.**
- **Enter an alerting name for the DN.**
- **Save changes.**



# Line Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows `https://192.168.1.10/ccmadmin/phoneSave.do`. The page title is "Cisco CallManager Administration" and it is logged in as "CCMAdministrator".

The "Phone Configuration" section is active, showing "Related Links: Back To Find/List". The "Status" section indicates "Update successful".

The "Association Information" section contains a list of configuration items:

- 1 **Line [1] - Add a new DN** (highlighted with a red box and a yellow callout bubble with the number 1)
- 2 Add a new BLF SD
- 3 Add a new BLF SD
- 4 Add a new BLF SD
- 5 Add a new BLF SD
- 6 Add a new SURL

Below this list is an "Add On Module(s)" section with options 7 through 13, all set to "None".

The "Device Information" section shows the following details:

- Product Type: Cisco 7960
- Device Protocol: SCCP
- Registration: Unknown
- IP Address: Unknown
- MAC Address\*: 0017ED2337A1
- Description: SEP0017ED2337A1
- Device Pool\*: SanJose
- Phone Button Template\*: 7960-1line-4blf-1service SCCP
- Softkey Template: < None >
- Common Phone Profile\*: Standard Common Phone Profile
- Calling Search Space: < None >
- AAR Calling Search Space: < None >

The bottom status bar shows "Done" and the IP address "192.168.1.10".

# Line Configuration

The screenshot displays the 'Directory Number Configuration' page in a Mozilla Firefox browser window. The address bar shows the URL: `https://192.168.1.10/ccadmin/directoryNumberEdit.do?devicekey=cfd25716-d219-fc5d-ea4f-b72504fd9a3c&index=1&mapkey=&newdn=2001`. The page title is 'Directory Number Configuration'. A 'Save changes' button is highlighted with a yellow callout box labeled '4'. Below the status bar, a 'Note' states: 'Changes to Line or Directory Number settings require a restart.' The 'Directory Number Information' section contains the following fields:

- Directory Number\*: 2001 (highlighted with a red box and callout '2')
- Route Partition: < None >
- Description: SJC 2001 (highlighted with a red box and callout '3')
- Alerting Name: Max Miller (highlighted with a red box and callout '3')
- ASCII Alerting Name: Max Miller
- Active

The 'Directory Number Settings' section includes several dropdown menus:

- Voice Mail Profile: < None > (Choose <None> to use system default)
- Calling Search Space: < None >
- Presence Group\*: Standard Presence group
- AAR Group: < None >
- User Hold Audio Source: < None >
- Network Hold Audio Source: < None >
- Auto Answer\*: Auto Answer Off

The 'Call Forward and Call Pickup Settings' section is partially visible at the bottom. The browser status bar shows 'Done' and the IP address '192.168.1.10'.

# Line Configuration

## Cisco Unified CallManager 5.0 line settings sections:

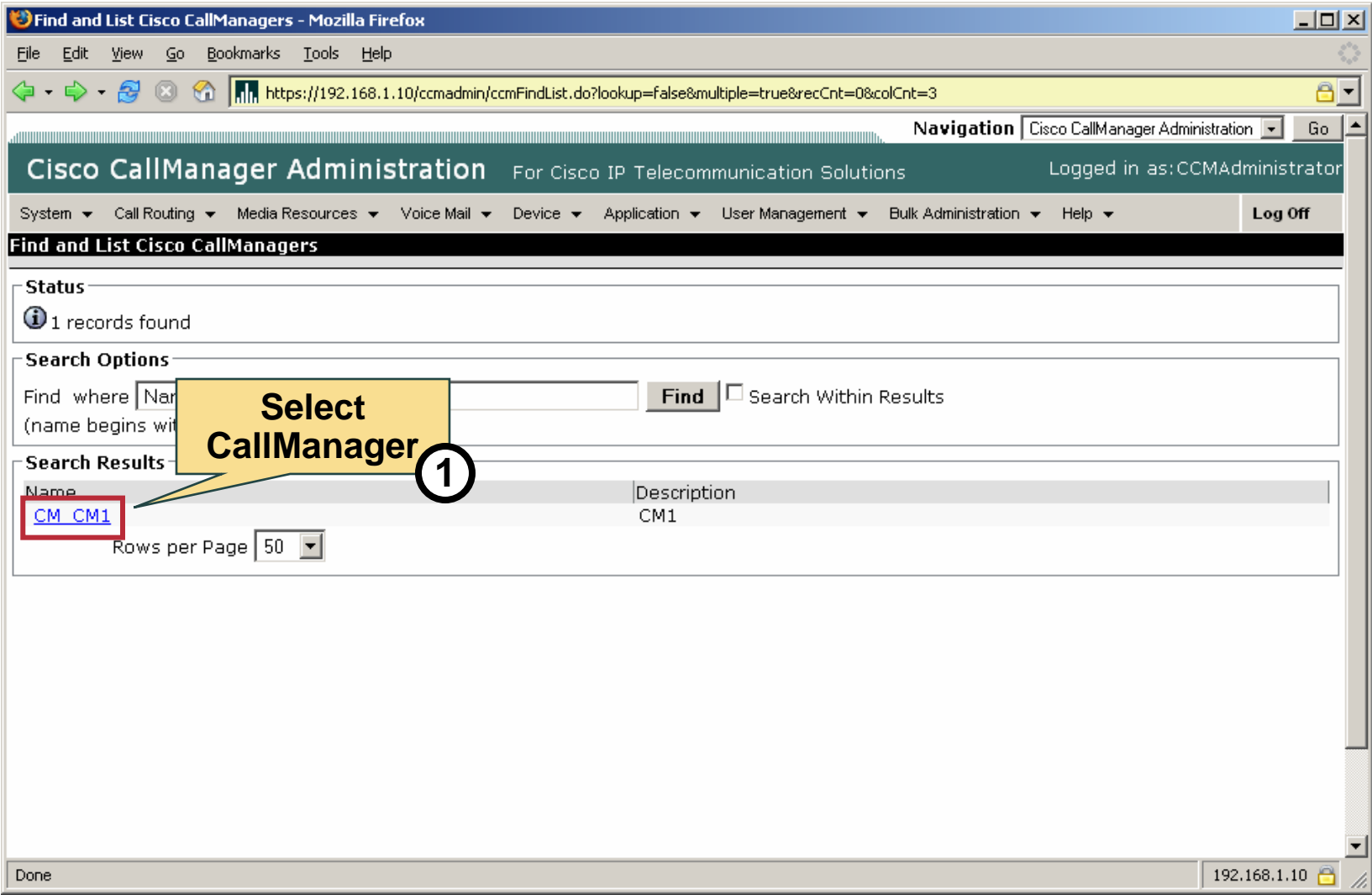
Section	Function
Directory Number Information	DN number configuration.
Directory Number Settings	DN settings configuration.
Call Forward and Call Pickup Settings	Call forward and pickup group allocation.
MLPP Alternate Party Settings	MLPP configuration.
Line on Device	Line specific phone configuration.
Multiple Call/Call Waiting Settings	Max. calls for line configuration.
Forwarded Call Information	Define information that should be forwarded.

# Configuring IP Phone Auto-Registration

**Auto-Registration can be used to easily add new phones to CallManager:**

- **Select the server on which auto-registration will be enabled.**
- **Enter a phone number range for auto-registered phones.**
- **Enable auto-registration on the specified Cisco Unified CallManager.**
- **Note: Manual re-configuration required to personalize auto-registered devices.**

# Configuring IP Phone Auto-Registration



# Configuring IP Phone Auto-Registration

The screenshot displays the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL: `https://192.168.1.10/ccmadmin/ccmEdit.do?key=848ec286-ac9a-4d14-b0ce-7c068385667c`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The configuration page is titled "Cisco CallManager Configuration" and includes a "Related Links" section with a "Back To Find/List" button. The main configuration area is divided into several sections:

- Status:** Shows "Status: Ready".
- Cisco CallManager:** Displays "CM\_CM1 (used by 12 devices)" and "CTI ID 1".
- Server Information:** Includes fields for "Cisco CallManager Server\*" (192.168.1.10), "Cisco CallManager Name\*" (CM\_CM1), and "Description" (CM1).
- Auto-registration Information:** Contains fields for "Starting Directory Number\*" (2900), "Ending Directory Number\*" (2999), "Partition" (< None >), and "External Phone Number Mask". A checkbox labeled "Auto-registration Disabled on this Cisco CallManager" is currently unchecked.
- Cisco CallManager TCP Port Settings for this Server:** Includes the "Ethernet Phone Port\*" set to 2000.

Two callout boxes provide instructions:

- Callout 2: "Enter DN range for auto-registration" points to the "Starting Directory Number" and "Ending Directory Number" fields.
- Callout 3: "Enable auto-registration" points to the "Auto-registration Disabled on this Cisco CallManager" checkbox.

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# Verify Endpoint Configuration

**To verify that the phone configuration is done successfully, do the following:**

- **Verify that the phone is registered.**
- **Verify the correct Cisco Unified CallManager is used.**
- **Verify the IP address of the phone.**
- **Verify that the lines are associated to the correct phone(s).**

# Verify Endpoint Configuration

The screenshot shows the Cisco CallManager Administration interface. The search results table is as follows:

	Device Name(Line)	Description	Device Pool	Device Protocol	Status	IP Address	Copy	Copy w/Lines
<input type="checkbox"/>	<a href="#">SEP00123F19CBD6</a>		<a href="#">Default</a>	SCCP	Registered with 192.168.1.10	<a href="#">10.128.128.88</a>		
<input type="checkbox"/>	<a href="#">SEP0017ED2337A1</a>	SEP0017ED2337A1	<a href="#">SanJose</a>	SCCP	Unknown	Unknown		

Annotations in the image:

- Verify that the phone is registered at the correct CallManager**: Points to the 'Status' column of the first row.
- Verify IP of the phone**: Points to the 'IP Address' column of the first row.
- Phone was never registered. Used wrong MAC?**: Points to the second row.



# Verify Endpoint Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL: `https://192.168.1.10/ccmadmin/directoryNumberEdit.do?key=66d90053-935f-ec72-c061-479153551f8a&mapkey=7288d028-a9eb-9d91-7a0b-be8775`. The page title is "Directory Number Configuration".

The interface includes a navigation menu with "Cisco CallManager Administration" and "Go" buttons. The user is logged in as "CCMAdministrator". The main menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". A "Log Off" button is also present.

The "Directory Number Configuration" section shows the following fields:

- Directory Number\*: 2001
- Route Partition: < None >
- Description: SJC 2001
- Alerting Name: Max Miller
- ASCII Alerting Name: Max Miller
- Allow Control of Device from CTI
- Associated Devices: SEP0017ED2337A1

A yellow callout box with the text "Verify that line is associated to the correct devices" points to the "Associated Devices" field. Below the field are buttons for "Edit Device" and "Edit Line Appearance". At the bottom of the field, there are "Dissociate Devices" controls.

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# Summary

- **Cisco Unified CallManager 5.0 has many functionalities that assist in configuring and managing SCCP IP phones.**
- **Device pool configuration is used to define global device settings for phones within that pool.**
- **IP phone button templates define how buttons on phones are used.**
- **To manually configure IP phones on Cisco Unified CallManager 5.0 it is necessary to know MAC address and type of the phone.**
- **At least one line needs to be configured for each phone.**
- **Auto-registration allows automatic registration of the phone and configuration of the line on Cisco Unified CallManager.**
- **Endpoint configuration can be verified by checking the registration status of the phone.**

# CISCO SYSTEMS





## **Deploying Cisco Unified CallManager 5.0 Endpoints**

# **Configuring SIP Endpoints in Cisco Unified CallManager 5.0**

# Objectives

- **SIP Phone Features**
- **Feature Issues**
- **Configuring Cisco SIP Phones**
- **Configuring 3rd Party SIP Phones**

# SIP Phone Features: 7905/7912 SIP Phones

**Minimum required features for 7905/7912 SIP phones to operate are available with software version 8.0:**

- **Registration redundancy.**
- **Failover to SRST.**
- **Reset/restart from CallManager.**
- **Configuration and local dial plan provisioned from Cisco Unified CallManager administration pages.**
- **Encrypted configuration files.**
- **Call preservation.**



# SIP Phone Features: 7940/7960 SIP Phones

**These required features for 7940/7960 SIP phones to operate are available with software version 8.0:**

- **Enhancement to release 7.5:**
  - **8.0 added better RFC compliance (3261 (SIP), 3264 (Offer Answer), and 3311(UPDATE)).**
- **Registration redundancy.**
- **Reset/Restart from CallManager.**
- **Failover to SRST.**
- **Configuration and local dial plan provisioned from CallManager administration pages.**
- **Encrypted Configuration files.**



# SIP Phone Features: 7940/7960 SIP Phones

**These required features for 7940/7960 SIP phones to operate are available with software version 8.0:**

- **Limited scope compared to the 7911/7941/7961/7970/7971 SIP implementation:**
  - **Music On Hold.**
  - **Display Name updates via Remote-Party-ID.**
- **End user feature interaction and UI operation is different between SCCP and SIP on the 7940/60 phones:**
  - **Look and Feel of existing firmware 7.5.**





# SIP Phone Features: 7911/7941/7961/7970/7971 SIP Phones

**SIP firmware 8.0 introduces advanced SIP support on the 7911/7941/7961/7970/7971 phone models:**

- Java architecture currently used in the SCCP phones was leveraged for its UI control and platform infrastructure.
- 7940/7960 SIP stack and call feature logic are ported and enhanced for the new phones.

**From an end-user perspective feature interaction and UI operation is nearly identical between SCCP and SIP:**

- 90% of the SCCP features have been ported to SIP on the 7911/7941/7961/7970/7971 phones.



# SIP Phone Features

## SIP vs SCCP configuration:

SIP	SCCP
Configuration file via TFTP	Configuration via TFTP
Softkey file via TFTP (Enhanced IP Phones only)	Softkey template via SCCP messages
Can have local dial plan	Does not have local dial plan
Registers via UDP, TCP or TLS	Registers via TCP or TLS

# SIP Phone Features

## Phone to CCM interaction comparison:

	Integrated with CTFTP	Sends MAC address	Downloads Softkey file	Downloads Dialplan file	Supports CCM failover / fallback	Supports Reset / Restart
Cisco 7905/12	Yes	Yes	No	Yes <sup>1</sup>	Yes	Yes
Cisco 7940/60	Yes	Yes	No	Yes	Yes	Yes
797x, 7941, 7961	Yes	Yes	Yes	Yes	Yes	Yes
3 <sup>rd</sup> -Party	No	No <sup>2</sup>	No	No	No <sup>2</sup>	No <sup>2</sup>

<sup>1</sup> Part of the configuration file.

<sup>2</sup> In the future, 3<sup>rd</sup>-party phones may add this support. Failover can be supported today via DNS SRV.

# Feature Issues

## What is **not** supported with the 7905/7912/7940/7960 SIP Phones?

- Anything requiring CMXML 3.1 (IPPM, IPMA, EM).
- Digit by digit dialing using Key Pad Markup Language.
- The 7914 sidecar module.
- Cisco VT Advantage (CVTA).
- CTI call control.
- Configurable Softkeys.
- Ad-Hoc Conference B2BUA mode (external conferencing resources).
- UI consistency with the SCCP phones.

# Configuring Cisco SIP Phones

**Configuration steps for Cisco SIP phones on Cisco Unified CallManager are similar to SCCP phone configuration:**

- **Add a new phone.**
- **Select the phone type.**
- **Select the device protocol SIP.**
- **Configure the device information.**
- **Add a new DN.**

# Configuring Cisco SIP Phones

The screenshot shows the 'Find and List Phones' interface in a Mozilla Firefox browser window. The browser's address bar displays the URL `https://192.168.1.10/ccmadmin/phoneFindList.do`. The page header includes 'Cisco CallManager Administration' and 'Logged in as: CCMAdministrator'. A navigation menu is visible with options like 'System', 'Call Routing', 'Media Resources', 'Voice Mail', 'Device', 'Application', 'User Management', 'Bulk Administration', and 'Help'. The main content area is titled 'Find and List Phones' and features a search section with the following elements:

- Search Options:** A form with 'Find Phone where' set to 'Device Name', a search type of 'begins with', a search input field, a 'Find' button, and a 'Search Within Results' checkbox.
- Search Results:** A message stating 'No active query. Please enter your search criteria using the options above.'
- Action:** An 'Add New' button is highlighted with a red box. A yellow callout bubble points to this button with the text 'Add a new phone' and a circled number '1'.

The status bar at the bottom of the browser window shows 'Done' on the left and '192.168.1.10' on the right.

# Configuring Cisco SIP Phones

The screenshot shows the 'Add a New Phone' page in the Cisco CallManager Administration interface. The page title is 'Add a New Phone' and the URL is 'https://192.168.1.10/ccmadmin/phoneEdit.do'. The interface includes a navigation menu with options like 'System', 'Call Routing', 'Media Resources', 'Voice Mail', 'Device', 'Application', 'User Management', 'Bulk Administration', and 'Help'. A 'Log Off' button is also present. The main content area is titled 'Add a New Phone' and features a 'Next' button and a dropdown menu for selecting a phone type. The dropdown menu is open, showing a list of phone models including 'Cisco 12 S', 'Cisco 12 SP', 'Cisco 12 SP+', 'Cisco 30 SP+', 'Cisco 30 VIP', 'Cisco 7902', 'Cisco 7905', 'Cisco 7910', 'Cisco 7911', 'Cisco 7912', 'Cisco 7920', 'Cisco 7935', 'Cisco 7936', 'Cisco 7940', 'Cisco 7941', 'Cisco 7941G-GE', 'Cisco 7960', and 'Cisco 7961'. A yellow callout bubble with the text 'Click next' and a circled '3' points to the 'Next' button. Another yellow callout bubble with the text 'Select phone type' and a circled '2' points to the dropdown menu.

# Configuring Cisco SIP Phones

Phone Configuration - Mozilla Firefox  
https://192.168.1.10/ccmadmin/phoneEdit.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Phone Configuration Related Links: Back To Find/List Go

Status  
Select the phone you would like to create  
Product Type Cisco 7961  
Select the device protocol SIP

Next

Click next 5

Select protocol 4

\*- indicates required item.  
- Device reset is not required for changes to Packet Capture Mode and Packet Capture Duration.

Done 192.168.1.10



# Configuring Cisco SIP Phones

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10/ccmadmin/phoneEdit.do`. The page title is "Cisco CallManager Administration" and it shows the user is logged in as "CCMAdministrator".

The "Phone Configuration" section is active, showing the following details:

- Status:** Status: Ready
- Phone Type:** Product Type: Cisco 7961, Device Protocol: SIP
- Device Information:**
  - MAC Address\*: 00082FA138E1
  - Description: SEP00082FA138E1
  - Device Pool\*: SanJose
  - Phone Button Template\*: Standard 7961 SIP
  - Softkey Template: < None >
  - Common Phone Profile\*: Standard Common Phone Profile
  - Calling Search Space: < None >
  - AAR Calling Search Space: < None >
  - Media Resource Group List: < None >
  - User Hold Audio Source: < None >

Three callout boxes with numbered circles (6, 7, 8) point to the MAC Address, Device Pool, and Phone Button Template fields respectively.

6 Enter MAC of phone

7 Select device pool

8 Select template

# Configuring Cisco SIP Phones

**Protocol Specific Information**

Packet Capture Mode*	None
Packet Capture Duration	0
Presence Group*	Standard Presence group
SIP Dial Rules	< None >
MTP Preferred Originating Codec*	711ulaw
SIP Phone Security Profile*	Standard SIP Profile for Auto Registration
Rerouting Calling Search Space	< None >
SUBSCRIBE Calling Search Space	< None >
SIP Profile*	Standard SIP Profile
Digest User	< None >

Media Termination Point Required  
 Unattended Port  
 Require DTMF Reception

**External Data Locations Information (Leave blank to use default)**

Information	
Directory	
Messages	
Services	
Authentication Server	
Proxy Server	
Idle	

Done

# Configuring Cisco SIP Phones

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows `https://192.168.1.10/ccmadmin/phoneSave.do`. The page title is "Cisco CallManager Administration" and it is logged in as "CCMAdministrator".

The "Phone Configuration" section is active, showing a status message: "Update failed. Could not insert new... INDEX column." Below this, the "Association Information" section is expanded, displaying a list of configuration options:

- 1 [Line \[1\] - Add a new DN](#) (highlighted with a red box and a yellow callout box labeled "11")
- 2 [Add a new SD](#)
- 3 [Add a new SURL](#)
- 4 [Add a new BLF SD](#)

The "Device Information" section shows the following details for a Cisco 7961 SIP phone:

- Product Type: Cisco 7961
- Device Protocol: SIP
- Registration: Unknown
- IP Address: Unknown
- MAC Address\*: 00082FA138E1
- Description: SEP00082FA138E1
- Device Pool\*: SanJose
- Phone Button Template\*: Standard 7961 SIP
- Softkey Template: < None >
- Common Phone Profile\*: Standard Common Phone Profile
- Calling Search Space: < None >
- AAR Calling Search Space: < None >

# Configuring Cisco SIP Phones

Directory Number Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/directoryNumberEdit.do?devicekey=dd1dbc24-47c5-82bc-e4cd-834453db9a41&index=1&mapkey=&newdn=2301

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Directory Number Configuration 13 Related Links: Back To Find/List Go

**Save changes**

Status

Status: Ready

Note: Changes to Line or Directory Number settings require a restart.

**Directory Number Information**

Directory Number\* 2301

Route Partition <None >

Description

Alerting Name

ASCII Alerting Name

Active

**Directory Number Settings**

Voice Mail Profile <None > (Choose <None> to use system default)

Calling Search Space <None >

Presence Group\* Standard Presence group

AAR Group <None >

User Hold Audio Source <None >

Done 192.168.1.10

**Enter DN** 12

**Slide 17**

---

**T. E.2** Table showing SIP configuration settings from CCM5.0 online help.  
Tolga Erdogan, 3/15/2006

# Configuring 3rd Party SIP Phones

## SIP Phone Configuration Sequence:

- **Configure an end user for the device.**
  - **Most SIP phones have an authorization ID that is used for digest-authentication.**
  - **Some SIP endpoints use the DN as the authorization ID, so the end user username has to be added as the DN.**
- **Configure the device.**
- **Associate the device with the end user.**
- **Configure the phone with an end user ID.**
  - **Device itself need to be provisioned via separate TFTP, HTTP, or phone configuration.**



# Configuring 3rd Party SIP Phones

**End User Configuration**

Status: Ready

**User Information**

LDAP Sync Status	Active
User ID*	2302
Password*	*****
Confirm Password*	*****
PIN*	*****
Confirm PIN*	*****
Last name*	Hunter
Middle name	
First name	Marc
Telephone Number	
Mail ID	
Manager User ID	
Department	

Enter user ID for 3rd party SIP device ②

Configure password and PIN ③

Enter name of user ④



# Configuring 3rd Party SIP Phones

The screenshot shows the 'Find and List Phones' interface in the Cisco CallManager Administration console. The browser window title is 'Find and List Phones - Mozilla Firefox' and the address bar shows 'https://192.168.1.10/ccmadmin/phoneFindList.do'. The page header includes 'Cisco CallManager Administration' and 'Logged in as: CCMAdministrator'. A navigation menu is visible with options like System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The main content area is titled 'Find and List Phones' and contains a search form. The search form has a 'Find Phone where' dropdown set to 'Device Name', a 'begins with' dropdown, a text input field, a 'Find' button, and a 'Search Within Results' checkbox. Below the search form is a 'Search Results' section with the message 'No active query. Please enter your search criteria using the options above.' and an 'Add New' button. A yellow callout box with the text 'Add a new phone' and a circled number '5' points to the 'Add New' button.

# Configuring 3rd Party SIP Phones

The screenshot shows the 'Add a New Phone' page in the Cisco CallManager Administration interface. The page title is 'Add a New Phone - Mozilla Firefox' and the URL is 'https://192.168.1.10/ccmadmin/phoneEdit.do'. The user is logged in as 'CCMAdministrator'. The page has a navigation menu with options like 'System', 'Call Routing', 'Media Resources', 'Voice Mail', 'Device', 'Application', 'User Management', 'Bulk Administration', and 'Help'. The main content area is titled 'Add a New Phone' and includes a 'Status' section showing 'Status: Ready'. Below this is a section titled 'Select the type of phone you would like to create' with a 'Phone Type\*' dropdown menu. The dropdown menu is open, showing a list of device types including 'Cisco 7920', 'Cisco 7935', 'Cisco 7936', 'Cisco 7940', 'Cisco 7941', 'Cisco 7941G-GE', 'Cisco 7960', 'Cisco 7961', 'Cisco 7961G-GE', 'Cisco 7970', 'Cisco 7971', 'Cisco 7985', 'Cisco ATA 186', 'Cisco IP Communicator', 'Cisco Unified Personal Communicator', 'H.323 Client', 'IP-STE', and 'Motorola CN622'. The last two options, 'Third-party SIP Device (Advanced)' and 'Third-party SIP Device (Basic)', are highlighted with a red box. A yellow callout box with the number 6 points to these options with the text 'Select SIP device type'. There are also informational icons and a 'Next' button on the page.

# Configuring 3rd Party SIP Phones

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10/ccmadmin/phoneEdit.do`. The page title is "Cisco CallManager Administration" and it shows the user is logged in as "CCMAdministrator".

The "Phone Configuration" section is active, showing the following configuration details:

- Status:** Ready
- Phone Type:** Product Type: Third-party SIP Device (Basic); Device Protocol: SIP
- Device Information:**
  - MAC Address\*: 000F1FBF7E92
  - Description: SEP000F1FBF7E92
  - Device Pool\*: SanJose
  - Phone Button Template\*: Third-party SIP Device (Basic)
  - Common Phone Profile\*: Standard Common Phone Profile
  - Calling Search Space: < None >
  - Media Resource Group List: < None >
  - Location\*: Hub\_None
  - Owner User ID: < None >

Three callouts are present:

- 7:** Enter MAC of phone (points to the MAC Address field)
- 8:** Select device pool (points to the Device Pool dropdown)
- 9:** Select template (points to the Phone Button Template dropdown)

# Configuring 3rd Party SIP Phones

The screenshot shows the 'Phone Configuration' web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10/ccmadmin/phoneEdit.do`. The interface includes several sections:

- Ignore Presentation Indicators (internal calls only)**:
- Protocol Specific Information**:
  - Presence Group\*: Standard Presence group
  - MTP Preferred Originating Codec\*: 711ulaw
  - SIP Phone Security Profile\*: Non Secure SIP Profile auth by String (highlighted with a red box and callout 10: "Select security profile")
  - Rerouting Calling Search Space: < None >
  - SUBSCRIBE Calling Search Space: < None >
  - SIP Profile\*: Standard SIP Profile (highlighted with a red box and callout 11: "Select standard SIP profile")
  - Digest User: 2302 (highlighted with a red box and callout 12: "Select configured user for device")
  - Media Termination Point Required
  - Unattended Port
- MLPP Information**:
  - MLPP Domain: < None >
- Secure Shell Information**:
  - Secure Shell User: [text input]
  - Secure Shell Password: [text input]

A 'Save' button is located at the bottom left. At the bottom, there are two informational messages:

- i** \*- indicates required item.
- i** - Device reset is not required for changes to Packet Capture Mode and Packet Capture Duration.

The status bar at the bottom shows 'Done' and the IP address '192.168.1.10'.

# Configuring 3rd Party SIP Phones

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows `https://192.168.1.10/ccmadmin/phoneSave.do`. The page title is "Phone Configuration - Mozilla Firefox". The navigation bar includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The main menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Phone Configuration" section is active, showing "Related Links: Back To Find/List".

The "Association Information" section contains a list of items:

- 1 [Line \[1\] - Add a new DN](#)
- 2 [Add a new SD](#)
- 3 Privacy
- 4 None

A yellow callout bubble with the text "Add a new DN for the phone" and a circled number "13" points to the first item in the list. The "Device Information" section shows the following configuration:

Product Type:	Third-party SIP Device (Basic)
Device Protocol:	SIP
Registration	Unknown
IP Address	Unknown
MAC Address*	000F1FBF7E92
Description	SEP000F1FBF7E92
Device Pool*	SanJose
Phone Button Template*	Third-party SIP Device (Basic)
Common Phone Profile*	Standard Common Phone Profile
Calling Search Space	< None >
Media Resource Group List	< None >
Location*	Hub_None

# Configuring 3rd Party SIP Phones

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays the URL: `https://192.168.1.10/ccmadmin/directoryNumberEdit.do?key=aea6acce-ec60-fe50e-5b8bab53b626&devicekey=571ed610-b19d-b22f-c43b-f1e6fc`. The page title is "Directory Number Configuration".

Navigation menu: Cisco CallManager Administration | Logged in as: CCMAdministrator | System | Resources | Voice Mail | Device | Application | User Management | Bulk Administration | Help | Log Off

Related Links: Configure Device (SEP000F1FBF7E92) | Go

**Save changes** (15)

**Enter DN** (14)

**Status**  
Status: Ready

Note: Changes to Line or Directory Number settings require restart.

**Directory Number Information**

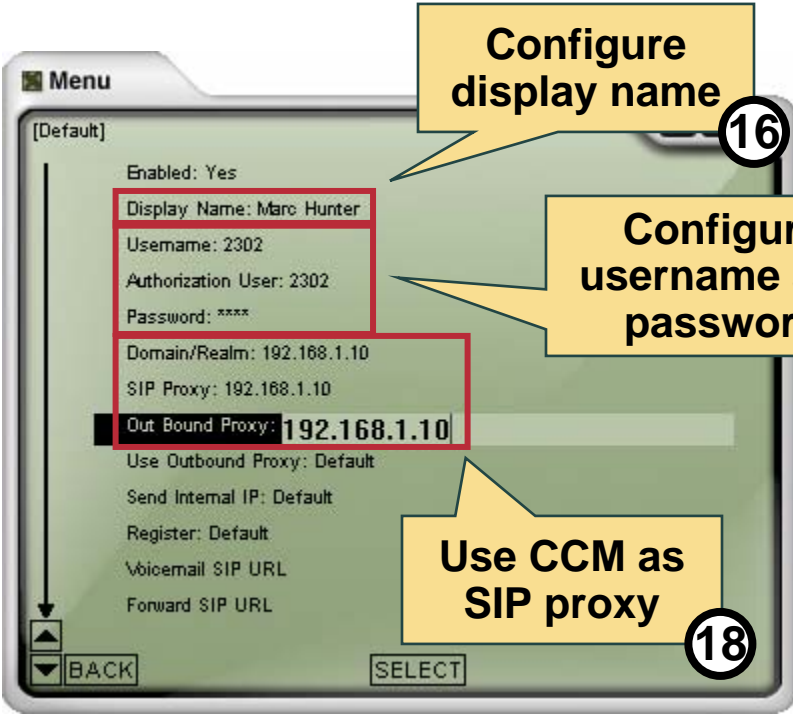
Directory Number*	2302
Route Partition	< None >
Description	
Alerting Name	
ASCII Alerting Name	
<input type="checkbox"/> Active	

**Directory Number Settings**

Voice Mail Profile	< None >	(Choose <None> to use system default)
Calling Search Space	Phones	
Presence Group*	Standard Presence group	
AAR Group	< None >	
User Hold Audio Source	< None >	

Done | 192.168.1.10

# Configuring 3rd Party SIP Phones



Menu  
[Default]

- Enabled: Yes
- Display Name: **Marc Hunter**
- Username: 2302
- Authorization User: 2302
- Password: \*\*\*\*
- Domain/Realm: 192.168.1.10
- SIP Proxy: 192.168.1.10
- Out Bound Proxy: 192.168.1.10**
- Use Outbound Proxy: Default
- Send Internal IP: Default
- Register: Default
- Voicemail SIP URL
- Forward SIP URL

BACK SELECT

**Configure display name** (16)

**Configure username and password** (17)

**Use CCM as SIP proxy** (18)



X-Lite

Logged in - Enter Phone Number

Your number is: 2302

Call Timer: 0:00:00

G711u G711a GSM iLBC

MUTE TRANSFER CONF

**Verify that phone is connected to CCM** (19)

**Make sure that CCM supported codecs configured** (20)

# Summary

- **Cisco Unified CallManager 5.0 supports many Cisco SIP phones as well as 3rd party SIP phones.**
- **SIP phone futures depend on the special Cisco SIP IP phone type.**
- **At the moment SIP phones do not support all of the same features as SCCP phones.**
- **Configuring Cisco SIP IP phones is similar to the configuration of Cisco SCCP IP phones.**
- **To configure 3rd party SIP IP phones, a user needs to be configured and associated to the device as well.**



# CISCO SYSTEMS





# Deploying Cisco Unified CallManager 5.0 Endpoints

## Managing Endpoints

# Objectives

- **Bulk Administration Tool (BAT) Overview**
- **Using BAT**
- **Configuring BAT Templates**
- **Creating CSV Files**
- **Adding Devices**
- **Scheduling Tasks**
- **Updating Devices**

# Bulk Administration Tool (BAT) Overview

**The Cisco Unified CallManager 5.0 Bulk Administration Tool (BAT) UI follows these guidelines:**

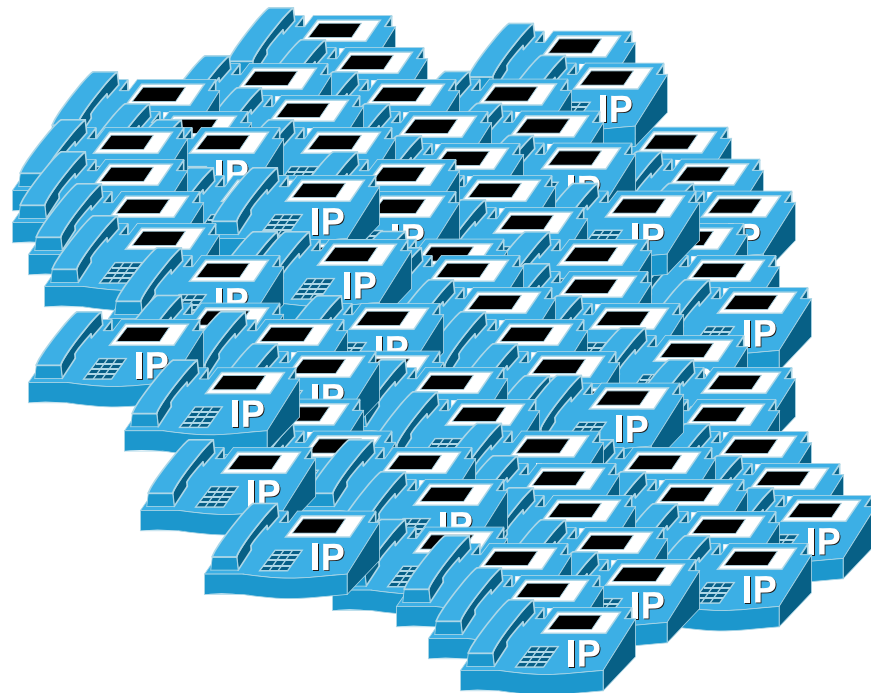
- **Integrated with the CCM Admin pages and is available by default (no plug-in required)**
- **All pages are available under the “Bulk Administration” menu**
- **Same look and feel as the CCM admin pages**
- **Supports Internationalization**
- **MLA support for BAT pages**
- **Tool for Auto-registered Phone Support (TAPS) is now part of Bulk Administration.**

# Bulk Administration Tool (BAT) Overview

The screenshot shows the Cisco CallManager Administration console in a Mozilla Firefox browser window. The browser title is "Cisco CallManager Console - Mozilla Firefox" and the address bar shows "https://192.168.1.10/ccmadmin/showHome.do". The page header includes "Cisco CallManager Administration For Cisco IP Telecommunication Solutions" and "Logged in as: CCMAdministrator". A navigation menu at the top contains "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", "Help", and "Log Off". The "Bulk Administration" menu is expanded, showing a list of options: "Upload/Download Files", "Phones", "Users", "Phones & Users", "Managers/Assistants", "User Device Profiles", "Gateways", "Forced Authorization Codes", "Client Matter Codes", "Call Pickup Group", "Job Scheduler", and "TAPS". Four yellow callout boxes with black text point to specific items in the menu: "Manage config files" points to "Upload/Download Files", "Manage devices" points to "Phones", "Schedule configuration" points to "Job Scheduler", and "Manage TAPS service" points to "TAPS". The page also contains a copyright notice for Cisco Systems, Inc. (1999-2005) and a disclaimer regarding cryptographic features.

# Using BAT

**BAT will allow management of many devices/records within a short period of time.**



# Using BAT

## **BAT administration contains the following parts:**

- **BAT templates are used to define general settings that fit all of the devices that should be added.**
- **CSV files are used to define device/record specific settings that should be bulk configured.**
- **Adding, updating, and deleting devices/records is done automatically based on query and CSVs.**
- **Additions, updates, and deletions can be scheduled to be performed at a defined time.**

# Using BAT

**BAT can be used on Cisco Unified CallManager 5.0 to add, update, and delete these devices and records:**

- **Cisco IP Phones**
- **Voice gateways (VG)**
- **CTI ports**
- **Users**
- **User Device Profiles**
- **Cisco IP Manager Assistant (IPMA) managers and assistants**
- **Ports on a Cisco Catalyst 6000 FXS Analog Interface Module**
- **Cisco VG200 series analog gateways and ports**
- **Forced Authorization Codes**
- **Client Matter Codes**
- **Call Pickup Groups**



# Using BAT

## **Bulk Provisioning Service (BPS):**

- **Bulk Provisioning Service is used by BAT to perform tasks.**
- **Bulk Provisioning Service is listed under Database services in the service activation pages.**
- **Service should be activated for scheduled jobs to be executed.**
- **Can be activated only on the publisher node in a cluster.**

# Using BAT

The screenshot shows the 'Service Activation' page in the Cisco CallManager Serviceability interface. The page is organized into several sections, each containing a table of services and their activation status. A yellow callout box highlights the 'Cisco Bulk Provisioning Service' in the 'Database and Admin Services' section, which is checked and activated.

Service Name	Activation Status
<input type="checkbox"/> Cisco IP Manager Assistant	Deactivated
<input checked="" type="checkbox"/> Cisco WebDialer Web Service	Activated

Service Name	Activation Status
<input type="checkbox"/> Cisco SOAP - CDRonDemand Service	Deactivated
<input checked="" type="checkbox"/> Cisco CAR Scheduler	Activated
<input checked="" type="checkbox"/> Cisco CAR Web Service	Activated

Service Name	Activation Status
<input type="checkbox"/> Cisco AXL Web Service	Deactivated
<input checked="" type="checkbox"/> Cisco Bulk Provisioning Service	Activated
<input type="checkbox"/> Cisco TAPS Service	Deactivated

Service Name	Activation Status
<input checked="" type="checkbox"/> Cisco Serviceability Reporter	Activated
<input type="checkbox"/> Cisco CallManager SNMP Service	Deactivated

Service Name	Activation Status
<input checked="" type="checkbox"/> Cisco CTL Provider	Activated

# Configuring BAT Phone Templates

**The BAT template configuration process is very similar to configuring each individual device or record:**

- **Add new device.**
- **Select device type.**
- **Configure device parameters.**

# Configuring BAT Phone Templates

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10/ccmadmin/bulkphonetemplateFindList.do`. The page title is "Find and List Phone Templates". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The main content area is titled "Find and List Phone Templates" and includes a "Related Links" section with a dropdown menu for "CAPF Report in File" and a "Go" button. Below this is a "+" icon and a "Status" section indicating "0 records found".

The "Search Options" section contains a search form with the following elements:

- Find Phone Template where:
- Search Within Results
- 

The "Search Results" section displays the message: "No active query. Please enter your search criteria using the options above." A red box highlights the "Add New" button, and a yellow callout bubble with the text "Add a new phone template" and a circled "1" points to it.

# Configuring BAT Phone Templates

The screenshot shows the 'Add a New Phone Template' page in the Cisco CallManager Administration interface. The page title is 'Add a New Phone Template' and the user is logged in as 'CCMAdministrator'. The page contains a form with a 'Status' field and a 'Phone Type' dropdown menu. The 'Phone Type' dropdown is open, showing a list of phone models including 'CTI Port', 'Cisco 12 S', 'Cisco 12 SP', 'Cisco 12 SP+', 'Cisco 30 SP+', 'Cisco 30 VIP', 'Cisco 7902', 'Cisco 7905', 'Cisco 7910', 'Cisco 7911', 'Cisco 7912', 'Cisco 7920', 'Cisco 7935', 'Cisco 7936', 'Cisco 7940', 'Cisco 7941', 'Cisco 7941 G-GE', 'Cisco 7960', and 'Cisco 7961'. A 'Next' button is visible below the dropdown. Annotations include a yellow callout box with the text 'Click next' and a circled '3' pointing to the 'Next' button, and another yellow callout box with the text 'Select phone type' and a circled '2' pointing to the dropdown menu.

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Add a New Phone Template Related Links: Back To Find/List Go

Status

Select the type of phone you would like to create

Phone Type\* - Not Selected -

Next

\*- indicates

Done

192.168.1.10

# Configuring BAT Phone Templates

The screenshot displays the Cisco CallManager Administration interface for configuring a phone template. The browser window title is "Phone Template Configuration - Mozilla Firefox" and the address bar shows "https://192.168.1.10/ccmadmin/bulkphonetemplateEdit.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". A navigation menu is visible with options like System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The main content area is titled "Phone Template Configuration" and includes a "Related Links" section with "Back To Find/List" and "Go".

The configuration form contains the following elements:

- Status:** A field with an information icon and the text "Status".
- Select the phone you will use to create:** A field with an information icon and the text "Select the phone you will use to create".
- Product Type:** A dropdown menu showing "Cisco 7961".
- Select the device protocol:** A dropdown menu with options "SCCP", "SCCP", and "SIP".
- Next:** A button highlighted with a red box.

Two callouts provide instructions:

- Callout 5: "Click next" pointing to the "Next" button.
- Callout 4: "Select protocol" pointing to the "Select the device protocol" dropdown menu.

A note at the bottom left states: "i \*- indicates required item." The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Configuring BAT Phone Templates

Phone Template Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/bulkphonetemplateEdit.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Phone Template Configuration Related Links: Back To Find/List Go

Status

Status: Ready

Phone Type

Product Type: Cisco 7960

Device Protocol: SCCP

Device Information

Template Name\* SJC\_7960

Description

Device Pool\* SanJose

Phone Button Template\* 7960-1line-4blf-1.service SCCP

Softkey Template < None >

Common Phone Profile\* Standard Common Phone Profile

Calling Search Space < None >

AAR Calling Search Space < None >

Media Resource Group List < None >

User Hold Audio Source < None >

Done 192.168.1.10

Enter name for phone template 6

Configure device parameters 7

# Configuring BAT Phone Templates

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10/ccmadmin/bulkphonetemplateSave.do`. The page title is "Phone Template Configuration" and the user is logged in as "CCMAdministrator".

The interface includes a navigation menu with options like "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". A "Log Off" button is also present.

The main content area is titled "Phone Template Configuration" and includes a "Related Links" section with a "Back To Find/List" link. Below this, there are icons for saving, deleting, and adding items.

The "Status" section shows "Add successful". The "Association Information" section has a "Modify Button Items" button and a list of items. Item 1 is highlighted with a red box and contains the text "Line [1] - Add a new DN". A yellow callout box with the text "Add a new line template for the phone template" and a circled number "8" points to this item. Other items in the list include "Add a new BLF SD" (items 2-5) and "Add a new SURF" (item 6). Below the list is an "Add On Module(s)" section with a table of modules.

The "Device Information" section contains the following fields:

- Device Protocol: SCCP
- Template Name\*: SJC\_7960
- Description: [Empty]
- Device Pool\*: SanJose
- Phone Button Template\*: 7960-1line-4blf-1service SCCP
- Softkey Template: < None >
- Common Phone Profile\*: Standard Common Phone Profile
- Calling Search Space: < None >
- AAR Calling Search Space: < None >
- Media Resource Group List: < None >
- User Hold Audio Source: < None >

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".



# Configuring BAT Phone Templates

The screenshot shows the Cisco CallManager Administration interface for configuring a line template. The browser window title is "Line Template Configuration - Mozilla Firefox". The URL is <https://192.168.1.10/ccmadmin/bulklinetemplateEdit.do?devicekey=1bf97459-4f7a-241d-f592-cf0292ad3ec0&index=1&routepartitionkey=35d43f3e-2>. The page is titled "Line Template Configuration" and shows the following fields:

- Status:** Status: Ready
- Directory Number Information:**
  - Line Template Name:  (highlighted with a red box and callout 9)
  - Route Partition: SJC-Phones
  - Description:
  - Alerting Name:
  - ASCII Alerting Name:
  - Active
- Directory Number Settings:**
  - Voice Mail Profile:  (Choose <None> to use system default)
  - Calling Search Space: SJC-Everywhere-CSS
  - Presence Group\*: Standard Presence group
  - AAR Group:
  - User Hold Audio Source:

Callout 9: Enter name for line template

Callout 10: Configure line parameters

# Creating CSV Files

**Cisco provides a template to create CSV files which have the special format that BAT requires:**

- **The template is a Microsoft Excel table using macros.**
- **The template can be personalized for the special needs of the environment.**
- **The macro in the template generates CSV files which have the correct format.**

# Creating CSV Files

The screenshot shows the Cisco CallManager Administration interface in a Mozilla Firefox browser window. The page title is "Find and List Files". The navigation bar includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The main content area is titled "Find and List Files" and contains a search form and a results table.

**Status:** 1 records found

**Search Options:** Find File where  Find  Search Within Results Using AND

**Search Results:**

File Name	Function Type
<input checked="" type="checkbox"/> bat.xlt	BAT Excel CSV Tool

Buttons: Add New, Select All, Clear All, Delete Selected, **Download Selected**, Rows per Page: 100

Annotations: "Select bat.xlt" (1) points to the search input; "Download" (2) points to the "Download Selected" button.

# Creating CSV Files

	A	B	C	D
1	<b>MAC Address/Device Name</b> (String[12/50] MANDATORY)	<b>Description</b> (String [50] OPTIONAL )	<b>Create File Format</b>	<b>Export to BAT Format</b>
2			<b>Select Model</b>	<input type="checkbox"/> <b>Dummy MAC Address</b>
3			<b>Phones</b>	<b>Maximum Number of Phone Lines</b> <input type="text" value="0"/>
4			<b>IP Port</b>	<b>Maximum Number of Speed Dials:</b> <input type="text" value="0"/>
5			<input type="radio"/> <b>IPES Client</b>	
6			<input type="radio"/> <b>VGC Virtual Phone</b>	<b>Maximum Number of IP Services and Parameters.</b> <input type="text" value="00"/>
7			<input type="radio"/> <b>VGC Phone</b>	<b>Note: Please enter the data in the &lt;Maximum number of IP Services&gt;:&lt;Maximum Number of Parameters&gt; format. For Eg:1:2</b>
8			<b>Note</b>	
9			If the Max number of calls and busy trigger are selected as line fields then busy trigger has to be less than or equal to the Max number of calls.	
10				
11				
12				
13				
14				
15				
16				
17				
18				
19				
20				
21				
22				

In Excel template create file format

3

# Creating CSV Files

**4 Add/remove fields**

**5 Order fields**

**6 Apply settings**

**7 Commit changes**

The screenshot shows a configuration window with the following sections:

- Device Fields:** A list of fields including AAR CSS, Built in Bridge, CSS, Device Pool, E.164, Extension Mobility, and Ignore Presentation Indicato. A red box highlights the right-side navigation buttons (>> and <<).
- Selected Device Fields:** A list containing MAC Address/Device Name and Description. A red box highlights the Up and Down ordering buttons.
- Line Fields:** A list of fields including Forward No Answer External, Forward No Answer Internal, Line Network Hold Audio Sour, Line Text Label, and Line User Hold Audio Source. A red box highlights the right-side navigation buttons (>> and <<).
- Selected Line Fields:** A list containing Directory Number, Line CSS (highlighted), Alerting Name, and External Phone Number Mask. A red box highlights the Up and Down ordering buttons.
- Create:** A button at the bottom of the main window, highlighted with a red box.
- Information Dialog:** A dialog box titled "Information" with the text "Do you wish to overwrite the existing CSV format?". The "Ja" button is highlighted with a red box.



# Creating CSV Files

E	F	G	H
<b>Alerting Name 1 (String [50] OPTIONAL )</b>	<b>External Phone Number Mask 1 (String [50] OPTIONAL )</b>	Create File Format	Export to BAT Format
Jimi Hendrix	1408555XXXX	Select Model	<input type="checkbox"/> Display MAC Address
Pete Townshend	1408555XXXX	<input checked="" type="radio"/> Phones	Maximum Number of Speed Dials: 0
Chuck Berry	1408555XXXX	<input type="radio"/> CTI Port	
Ian Gillan	1408555XXXX	<input type="radio"/> H.323 Client	
Keith Emerson	1408555XXXX		
John Lennon	1408555XXXX		

**Cisco CallManager Bulk Administration Tool**

Enter file path here:

C:\xlsDataFiles\Phones-02152006134330

Browse OK Cancel

**Microsoft Excel**

Data successfully exported to C:\xlsDataFiles\Phones-02152006134330.txt.

OK

**Export to CSV** 9

**Save changes** 10

Note: Please enter the data in the <Maximum number of IP Services>:<Maximum Number of Parameters> format. For Eg:1:2

# Adding Devices

**The following tasks need to be performed to add devices or records using BAT:**

- **Upload CSV to Cisco Unified CallManager 5.0**
- **Validate file content.**
- **Select template**
- **Run “Insert Task”.**
- **Verify that the insert is successful.**



# Adding Devices

The screenshot shows the Cisco CallManager Administration web interface. The browser window is titled "Find and List Files - Mozilla Firefox" and the address bar shows "https://192.168.1.10/ccmadmin/bulkfileuploadFindList.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Find and List Files" section has a status bar indicating "1 records found". The search options include "Find File where" with a dropdown menu set to "Name" and "begins with" set to an empty field. The search results table has the following content:

File Name	Function Type
bat.xlt	BAT Excel CSV Tool

Below the table are buttons for "Add New", "Select All", "Clear All", "Delete Selected", and "Download Selected". The "Add New" button is highlighted with a red box. A callout bubble with a yellow background and a black border points to the "Add New" button, containing the text "Add new file to BAT" and a circled "1".

# Adding Devices

The screenshot shows the 'File Upload Configuration' page in the Cisco CallManager Administration interface. The page is titled 'File Upload Configuration' and includes a 'Status' section showing 'Ready'. The main section is 'Upload the CSV file', which contains a 'File: \*' field with the path 'C:\xlsDataFiles\Phones-02152006134330.txt' and a 'Browse...' button. Below the file field is a 'Select The Target \*' dropdown menu with a list of options: Phones, - Not Selected -, Phones, Users, UDP, VG200 Gateways, CAT6k FXS Ports, Pickup Groups, Forced Authorization Codes, Client Matter Codes, Phones/users, Managers, Assistants, and Managers Assistants. The 'Phones' option is selected. A 'Save' button is located below the dropdown. Annotations include a yellow callout box labeled 'Enter file to upload' with a circled '2' pointing to the file field, and another yellow callout box labeled 'Select target' with a circled '3' pointing to the dropdown menu. The browser window title is 'File Upload Configuration - Mozilla Firefox' and the address bar shows 'https://192.168.1.10/ccmadmin/bulkfileuploadEdit.do'. The page footer includes 'Done' and the IP address '192.168.1.10'.

# Adding Devices

The screenshot shows the Cisco CallManager Administration interface in a Mozilla Firefox browser window. The page title is "File Upload Configuration". The status is "Ready". The "Upload the CSV file" section contains the following fields:

- File: \* C:\xlsDataFiles\Phones-02152006134330.txt
- Select The Target \* Phones
- Select Transaction Type \* (dropdown menu open)
- Overwrite File if it exists

The dropdown menu for "Select Transaction Type" is open, showing the following options:

- Insert Phones - Specific Details (highlighted)
- Not Selected -
- Insert Phones - All Details
- Delete Phones - Custom File
- Update Phones - Custom File
- Phone Add Lines
- Reset/Restart Phones Custom File
- Update Phones - CSV File

A yellow callout box with a speech bubble points to the dropdown menu, containing the text: "Select transaction type" with a circled number "4".

At the bottom of the browser window, the status bar shows "Done" and the IP address "192.168.1.10".

# Adding Devices

**Validate Phones Configuration**

Status  
Status: Ready

**Validate Phones**

Validate Phones Specific Details

File Name \*  [\(View File\)](#)

Phone Template Name \*  [\(View File\)](#)

Validate Phones All Details

File Name \*  [\(View File\)](#)

\* - indicates a required field.

**5** Select file to validate

**6** Select template that should be used

**7** Start Validation

# Adding Devices

The screenshot shows the Cisco CallManager Administration web interface. The page title is "Find and List Jobs". The status indicates "1 records found". The search options are set to "Find Job where User" and "begins with". The search results table shows one job with ID 1140008933, scheduled on 02/15/2006 05:08:53, with a status of "Completed".

**Search for completed jobs** (8)

**Select validation** (9)

Job Id	Scheduled Date Time	Submit Date Time	Sequence	Description	Status	Last User
<a href="#">1140008933</a>	02/15/2006 05:08:53	02/15/2006 05:08:53	0	Validate Specific Phones	Completed	CCMAdmin

# Adding Devices

**Job Configuration** - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/bulkjobEdit.do?key=79385643-c0b3-8c68-c695-3216ffccc3a5

Job Configuration Related Links: Back To Find/List Go

**Status**

Status: Ready

Server Date and Time: February 15, 2006 05:34:22 PST

**Job Details**

Job id\* 1140008933  
Job Status\* Completed  
Scheduled Date Time 02/15/2006 05:08:53  
Submit Date Time 02/15/2006 05:08:53  
Sequence\* 0  
Job Description Validate Specific Phones  
Frequency\* Once  
Job End Time  
Last Modified By CCAdministrator

**Transaction Details**

CSV File Name [Phones-02152006134330.txt](#)  
Template Name [SJC\\_7960](#)

**Job Results**

Job Launched Date Time	Job Result Status	Number Of Records Processed	Number Of Records Failed	Total Number Of Records	Log File Name.
02/15/2006 05:08:54	Success	6	0	6	<a href="#">1140008933#02152006050854.txt</a>

Delete

Done 192.168.1.10

**Verify that all records are processed** ⑩

# Adding Devices

The screenshot shows the 'Insert Phones Configuration' page in the Cisco CallManager Administration interface. The page is titled 'Insert Phones Configuration' and is part of the 'Cisco CallManager Administration' system. The user is logged in as 'CCMAdministrator'. The page contains several sections:

- Status:** Shows 'Status: Ready'.
- Insert Phones:** Contains two radio button options: 'Insert Phones Specific Details' (selected) and 'Insert Phones All Details'. Under 'Insert Phones Specific Details', there are two dropdown menus: 'File Name \*' (set to 'Phones-02152006134330.txt') and 'Phone Template Name \*' (set to 'SJC\_7960'). There are also links for '(View File)' and '(View Sample File)'. A checkbox 'Create Dummy MAC Address (For CTI Port, Create Dummy Device Name)' is present and unchecked.
- Job Information:** Contains a text field for 'Job Description' (set to 'Insert Phones - Specific Details') and two radio button options: 'Run Immediately' (selected) and 'Run Later (To schedule and activate this job, use Job Configuration page.)'.
- Submit:** A button labeled 'Submit' is located at the bottom left of the form.

Three callout boxes with numbers in circles provide instructions:

- 11:** 'Select file and template' points to the 'File Name' and 'Phone Template Name' dropdown menus.
- 12:** 'Select immediately' points to the 'Run Immediately' radio button.
- 13:** 'Start adding' points to the 'Submit' button.

# Adding Devices

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays the URL: `https://192.168.1.10/ccmadmin/bulkjobFindList.do?key=f24d8a64-e59c-8207-a01f-da0d77d7ceab`. The page title is "Find and List Jobs".

The interface includes a navigation menu with options: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The user is logged in as "CCMA".

The "Find and List Jobs" section shows a status of "2 records found" and a server date and time of "February 15, 2006 05:32:40 PST". Search options include a search box, a "Find" button, and a checkbox for "Search Within Results Using".

The "Search Results" table displays the following data:

Job Id	Scheduled Date Time	Submit Date Time	Sequence	Description	Status	Last User
<a href="#">1140009858</a>	02/15/2006 05:24:18	02/15/2006 05:24:18	1	Insert Phones - Specific Details	Completed	CCMAdmin
<a href="#">1140008933</a>	02/15/2006 05:08:53	02/15/2006 05:08:53	0	Validate Specific Phones	Completed	CCMAdmin

Below the table are buttons for "Select All", "Clear All", "Delete Selected", "Activate Selected", and "Stop Processing". The "Rows per Page" is set to 100.

A yellow callout box with the text "Verify a successful insertion" and a circled number "14" points to the first row of the search results table.



# Scheduling Tasks

**The following tasks need to be performed to schedule BAT tasks on Cisco Unified CallManager:**

- **Create a task that needs to have a scheduled time.**
- **Configure the time to start task.**

# Scheduling Tasks

The screenshot shows the 'Insert Phones Configuration' page in the Cisco CallManager Administration interface. The page is titled 'Insert Phones Configuration' and is part of the 'Cisco CallManager Administration' system. The user is logged in as 'CCMAdministrator'. The page contains several sections:

- Status:** Shows 'Status: Ready'.
- Insert Phones:** Contains two radio button options:
  - Insert Phones Specific Details:** Selected. Includes fields for 'File Name \*' (Phones-02152006134330.txt) and 'Phone Template Name \*' (SJC\_7960). There are links for '(View File)' and '(View Sample File)'. A checkbox for 'Create Dummy MAC Address (For CTI Port, Create Dummy Device Name)' is present.
  - Insert Phones All Details:** Not selected. Includes a 'File Name' field set to '- Not Selected -' and links for '(View File)' and '(View Sample File)'. A checkbox for 'Override the existing configuration' is present.
- Job Information:** Includes a 'Job Description' field (Insert Phones - Specific Details) and two radio button options:
  - Run Immediately:** Not selected.
  - Run Later (To schedule and activate this job, use Job Configuration page.):** Selected.
- Submit:** A button at the bottom left of the form.

Two callout boxes with numbers 1 and 2 provide instructions:

- 1:** Run configured task later (pointing to the 'Run Later' radio button).
- 2:** Create Job (pointing to the 'Submit' button).

# Scheduling Tasks

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10/ccmadmin/bulkjobFindList.do`. The page title is "Find and List Jobs".

The interface includes a navigation menu with options like "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help".

The "Find and List Jobs" section shows a status of "1 records found". The server date and time is "February 15, 2006 05:27:08 PST".

The search options are set to "Find Job where" "User" "begins with" "Completed Jobs" and "Find".

The search results table is as follows:

Job Id	Scheduled Date Time	Submit Date Time	Sequence	Description	Status	Last User
<input type="checkbox"/> <a href="#">1140010019</a>	02/15/2006 05:26:59	02/15/2006 05:26:59	10	Insert Phones - Specific Details	Hold	CCMAdmin

Below the table are buttons for "Select All", "Clear All", "Delete Selected", "Activate Selected", and "Stop Processing". The "Rows per Page" is set to 100.

A yellow callout bubble with the text "Select scheduled task" and a circled "3" points to the first search result.

# Scheduling Tasks

The screenshot shows the Cisco CallManager Administration interface in Mozilla Firefox. The page title is "Job Configuration" and the user is logged in as "CCMAdministrator". The "Job Details" section contains the following information:

Job id*	1140010019
Job Status*	Hold
Scheduled Date Time*	Feb 15 2006 5:30:00
Submit Date Time	02/15/2006 05:26:59
Sequence*	10
Job Description	Insert Phones - Specific Details
Frequency*	Once
Job End Time	
Last Modified By	CCMAdministrator

The "Transaction Details" section shows:

CSV File Name	<a href="#">Phones-02152006134330.txt</a>
Template Name	<a href="#">SJC_7960</a>

At the bottom, there are buttons for "Save", "Delete", and "Activate Job".

Annotations:

- Callout 4: "Configure start time for scheduled task" points to the "Scheduled Date Time" field.
- Callout 5: "Activate task" points to the "Activate Job" button.

# Updating Devices

**There are two ways to update devices/records with BAT:**

- **Use a CSV file containing the updated information:**
  - The process used is similar to adding a new device.
- **Search for a group of devices and change all of the devices or records within that selection to the same settings:**
  - Execute a search for the devices or records to update.
  - Select fields that should be updated on the specified devices.
  - Click the Create Task button.
- **Useful for updating parameters for a large number of devices.**

# Updating Devices

**Define devices that should be updated** ①

**Verify selection** ②

**Update Phones Query**

Status  
6 records found

**Search Options**

Find Phone where Description begins with SJC Find Search Within Results Using AND  
(device.description begins with SJC)

**Search Results**

	Device Name(Line)	Description	Device Pool	Device Protocol	Status	IP Address
7960	SEP000173AF21BE	SJC 2600	SanJose	SCCP	Unknown	Unknown
7960	SEP000173AF21BF	SJC 2601	SanJose	SCCP	Unknown	Unknown
7960	SEP000173AF21C1	SJC 2602	SanJose	SCCP	Unknown	Unknown
7960	SEP000173AF21C2	SJC 2603	SanJose	SCCP	Unknown	Unknown
7960	SEP000173AF21C3	SJC 2604	SanJose	SCCP	Unknown	Unknown
7960	SEP000173AF21C4	SJC 2605	SanJose	SCCP	Unknown	Unknown

Next Rows per Page 100

# Updating Devices

The screenshot shows the 'Update Phones' web interface in Mozilla Firefox. The interface includes a status section, a 'Reset/Restart Phones' section with radio buttons, and a 'Device Information' section with a list of fields and their target values. Three callouts are present: callout 3 points to the 'Device Information' section, callout 4 points to the 'Reset/Restart Phones' section, and callout 5 points to the 'Create task' button.

**5 Create task**

**4 Define if phones should be reset/restarted after update**

**3 Select fields that should be updated and target value**

**Status**  
Status: Ready

**Reset/Restart Phones**  
 Don't Reset/Restart phones  Reset phones  Restart phones

**Device Information**

<input type="checkbox"/> Description	
<input checked="" type="checkbox"/> Device Pool*	SanJose
<input checked="" type="checkbox"/> Phone Button Template*	7960-1line-5blf SCCP
<input type="checkbox"/> Softkey Template	< None >
<input type="checkbox"/> Calling Search Space	< None >
<input type="checkbox"/> AAR Calling Search Space	< None >
<input type="checkbox"/> Media Resource Group List	< None >
<input type="checkbox"/> User Hold Audio Source	< None >
<input type="checkbox"/> Network Hold Audio Source	< None >
<input checked="" type="checkbox"/> Location*	SJC
<input type="checkbox"/> User Locale	< None >
<input type="checkbox"/> Network Locale	< None >
<input type="checkbox"/> Built In Bridge*	< None >
<input type="checkbox"/> Privacy*	< None >

Done 192.168.1.10

# Summary

- **The Bulk Administration Tool (BAT) is fully integrated with Cisco Unified CallManager 5.0 Administration.**
- **BAT can be used to add, update, and delete an enormous number of devices or records within a short period of time.**
- **BAT templates define basic device configuration and are configured similar to each individual device.**
- **To create CSV files that meet Cisco Unified CallManager 5.0 requirements a Microsoft Excel BAT template is used.**
- **To add devices it is necessary to use the correct BAT template and CSV file.**
- **BAT tasks can be performed at a specified time using the Task Schedule option.**
- **Either a CSV file or device selection can be used to update devices.**



# CISCO SYSTEMS





## Deploying a Dial Plan on CallManager 5.0

# Understanding Dial Plans

# Objectives

- **Dial Plan Overview**
- **Endpoint Addressing**
- **Call Routing and Path Selection**
- **Digit Manipulation**
- **Call Privileges**
- **Call Coverage**

# Dial Plan Overview

**A dial plan defines how calls are interconnected:**

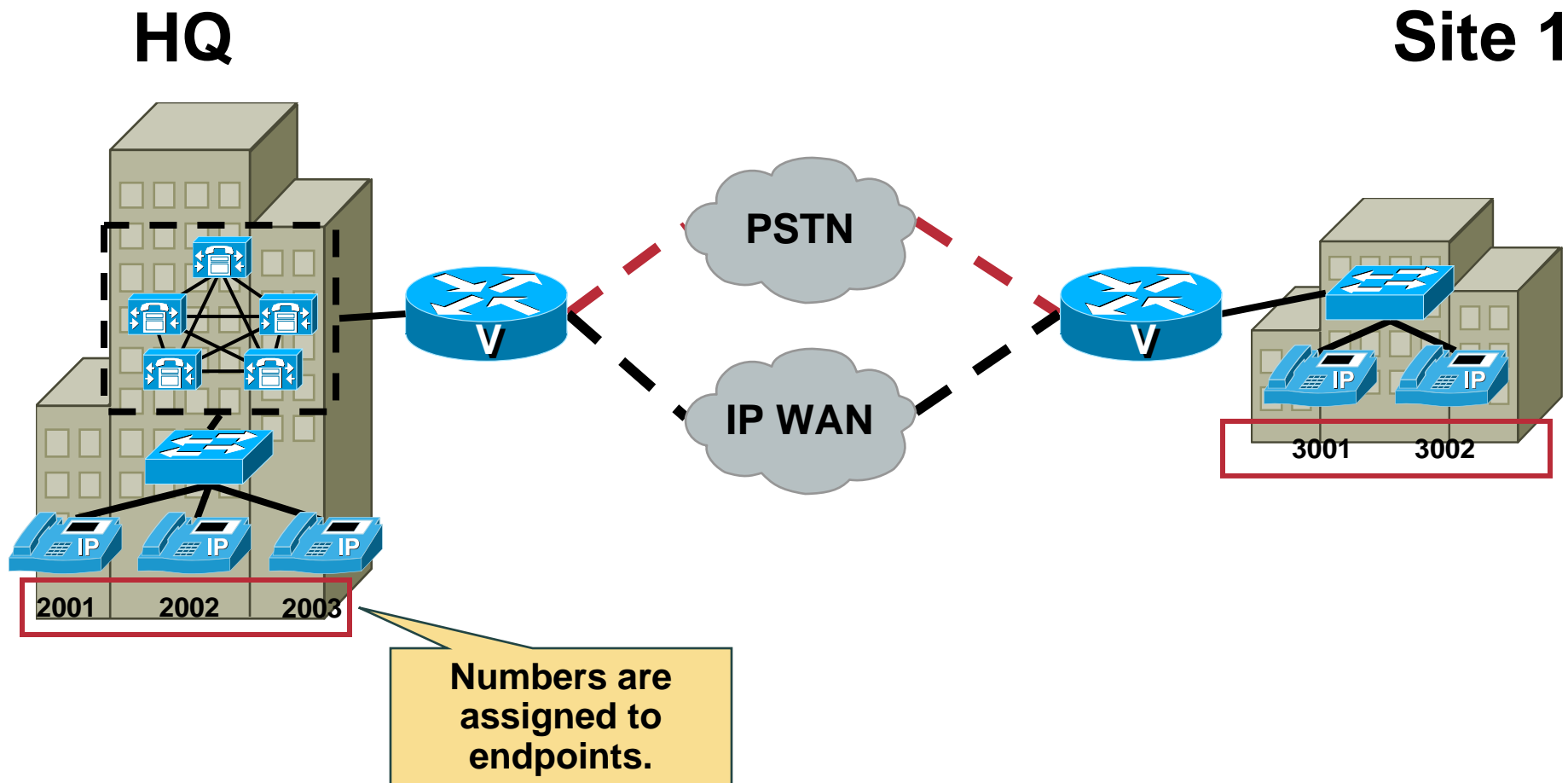
- **Directory numbers (DNs) used by endpoints.**
- **Where to and how calls are routed.**
- **How digits are modified.**
- **Restrict the destinations a user can dial.**
- **Ensure that incoming are answered.**

# End Point Addressing

## End Point Addressing has the following characteristics:

- **Directory numbers are assigned to end points, e.g. phones.**
- **Internal extensions are mapped to inbound PSTN calls.**
  - **Often dependant on range of DID numbers.**
- **The biggest challenge: creating an end point addressing scheme in multi-site environments.**

# End Point Addressing

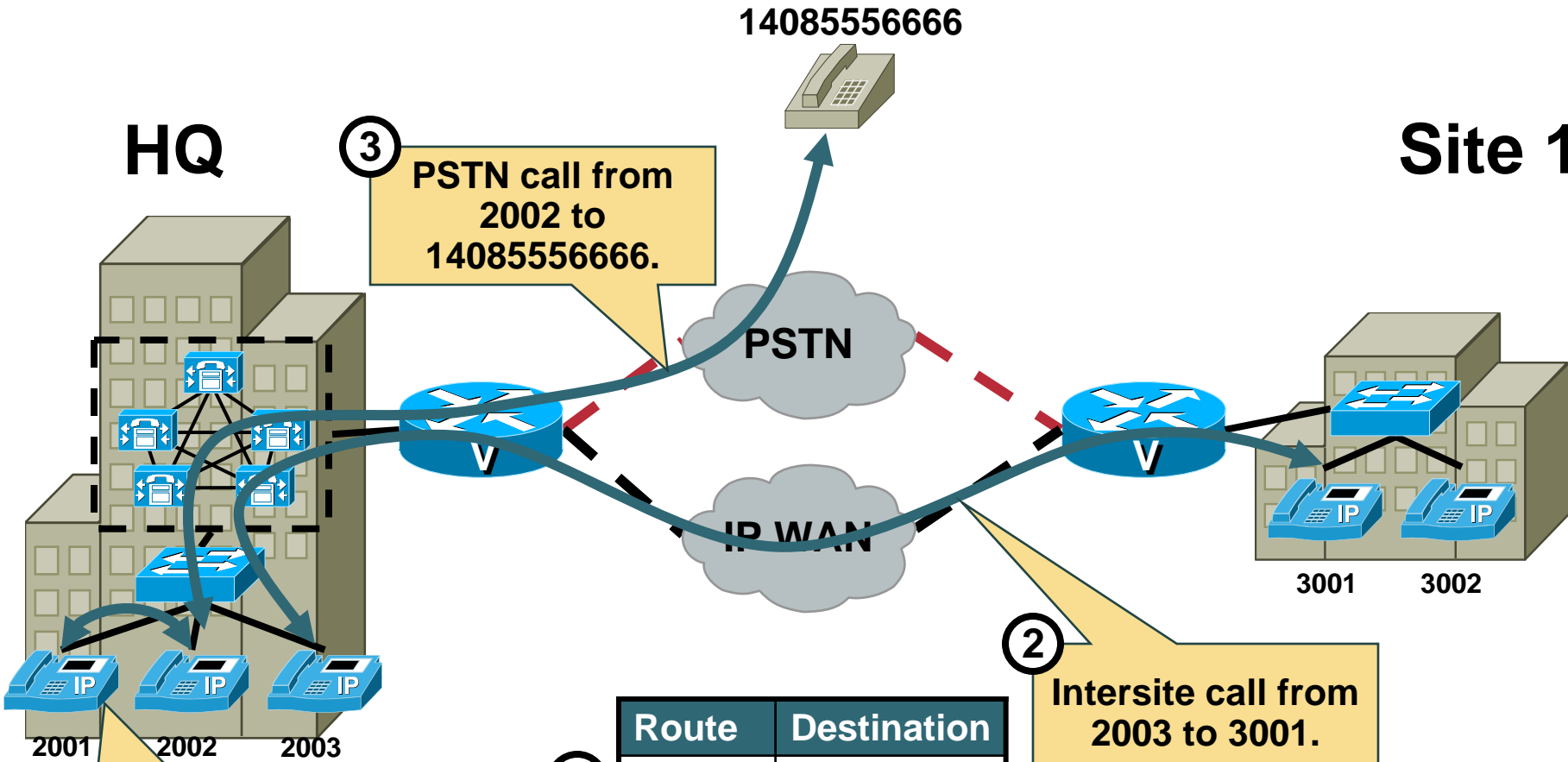


# Call Routing and Path Selection

**Call Routing and Path Selection are essential for telephony systems:**

- **Route the call depending on the dialed number.**
  - Very similar to destination based IP routing.
- **Differentiate between:**
  - Intrasite Routing
  - Intersite Routing
  - PSTN Routing
- **Select the appropriate path:**
  - IP vs. POTS path.
  - Overflow routing.
  - Time of Day routing.

# Call Routing and Path Selection



① Intrasite call from 2001 to 2002.

- ①
- ②
- ③

Route	Destination
2XXX	HQ
3XXX	Site 1
9!	PSTN

② Intersite call from 2003 to 3001.

③ PSTN call from 2002 to 14085556666.

Site 1

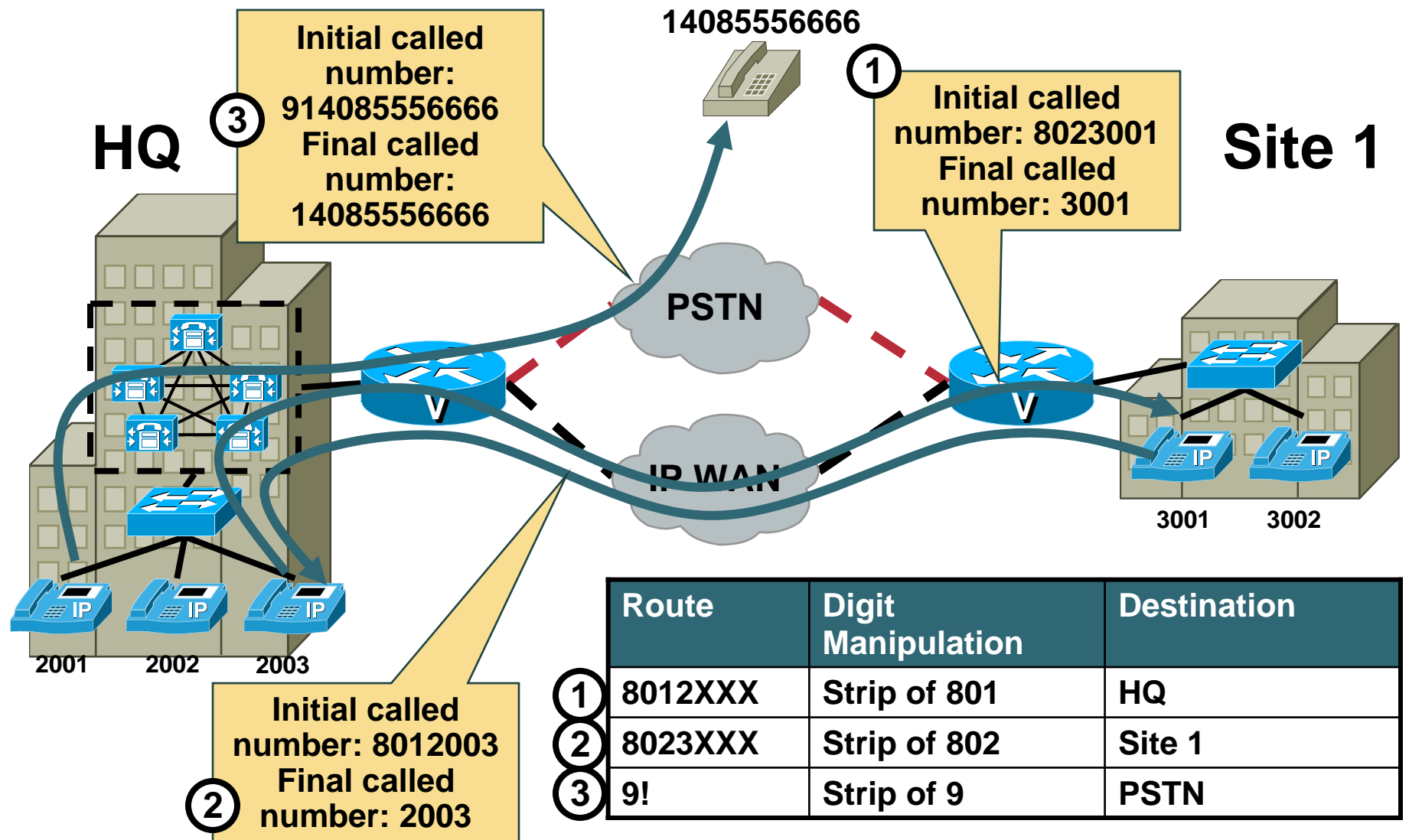


# Digit Manipulation

**Digit Manipulation is closely connected to call routing:**

- **Correct calling and called party number presentation.**
- **Required for site code dialing and short dials.**
- **Can solve overlapping end point DN issues.**

# Digit Manipulation



# Call Privileges

**Calling Privileges define the destinations a user can call:**

- **Used to control telephony charges.**
  - **Block costly service numbers.**
  - **Restrict international calls.**
- **Often called “Class of Service” in PBX systems**
  - **Do not confuse with Layer 2 COS**
- **Defined classes are assigned to individual users or user groups.**

# Call Privileges

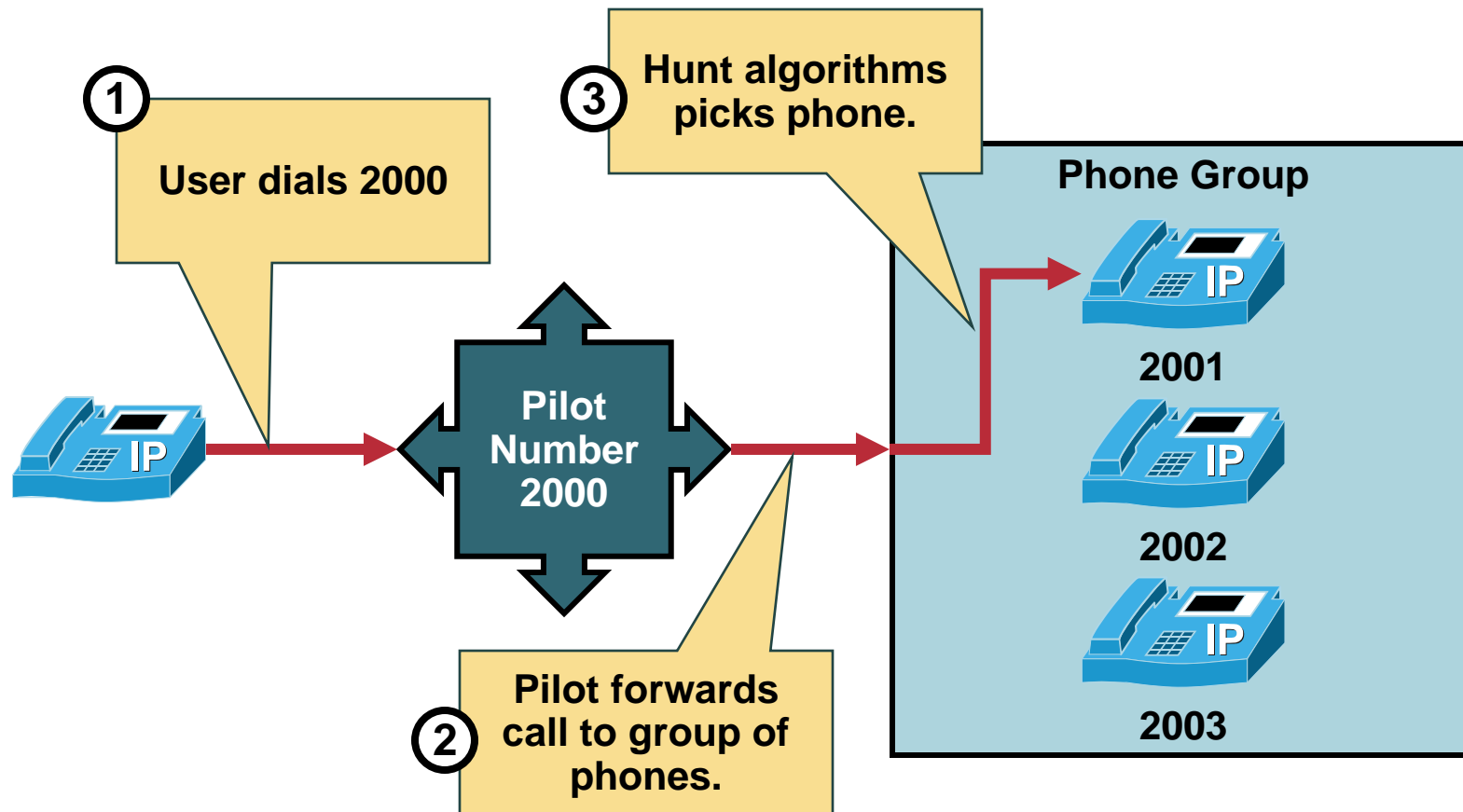
Calling Privilege Class (Class of Service)	Allowed Destinations
<b>Internal</b>	<ul style="list-style-type: none"><li>• Internal</li><li>• Emergency</li></ul>
<b>Local</b>	<ul style="list-style-type: none"><li>• Internal</li><li>• Emergency</li><li>• Local PSTN</li></ul>
<b>Long Distance</b>	<ul style="list-style-type: none"><li>• Internal</li><li>• Emergency</li><li>• Local PSTN</li><li>• Long Distance PSTN</li></ul>
<b>International</b>	<ul style="list-style-type: none"><li>• Internal</li><li>• Emergency</li><li>• Local PSTN</li><li>• Long Distance PSTN</li><li>• International PSTN</li></ul>

# Call Coverage

**Call Coverage ensures that all incoming calls are answered:**

- **Used for individuals:**
  - Ring other phones if original called phone is not answering.
- **Used for user groups with pilot numbers:**
  - Hunt through multiple phones
  - Ring multiple phones

# Call Coverage



# Summary

- **A dial plan defines how calls are interconnected.**
- **Endpoint addressing defines the DNs assigned to end points.**
- **Call routing and path selection determine where and how to route a call.**
- **Digit manipulation modifies calling and called party numbers.**
- **Call privileges enable or restrict users to reach certain destinations.**
- **Call coverage ensures that all incoming calls are answered.**

# CISCO SYSTEMS







**Deploying a Dial Plan on CallManager 5.0**

# **Understanding Cisco Unified CallManager Dial Plan Components**

# Objectives

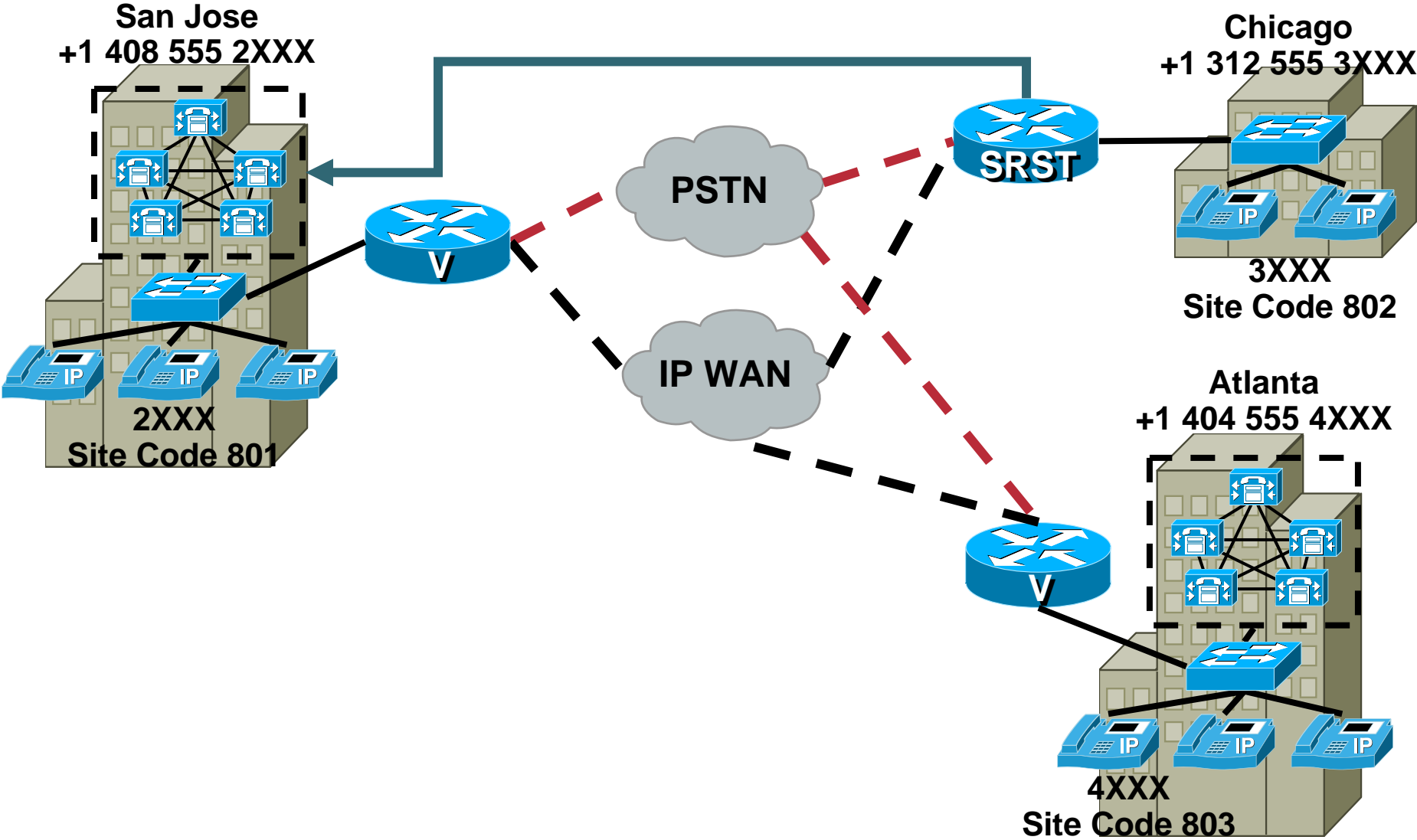
- **Cisco Unified CallManager Dial Plan Components Overview**
- **Partitions and Calling Search Spaces**
- **On/Off Cluster Routing**
- **Route Patterns**
- **Route Lists and Route Groups**
- **Translation Patterns**
- **Line Groups, Hunt Lists and Hunt Pilots**
- **Review of Cisco Unified CallManager Dial Plan Components**

# Cisco Unified CallManager Dial Plan Components Overview

## **CCM has the following dial plan components:**

- **Directory Numbers**
- **Partitions**
- **Calling Search Spaces**
- **(SIP) Route Patterns**
- **Route Groups and Route Lists**
- **Translation Patterns**
- **Line Group, Hunt List and Hunt Pilot**

# Cisco Unified CallManager Dial Plan Components Overview

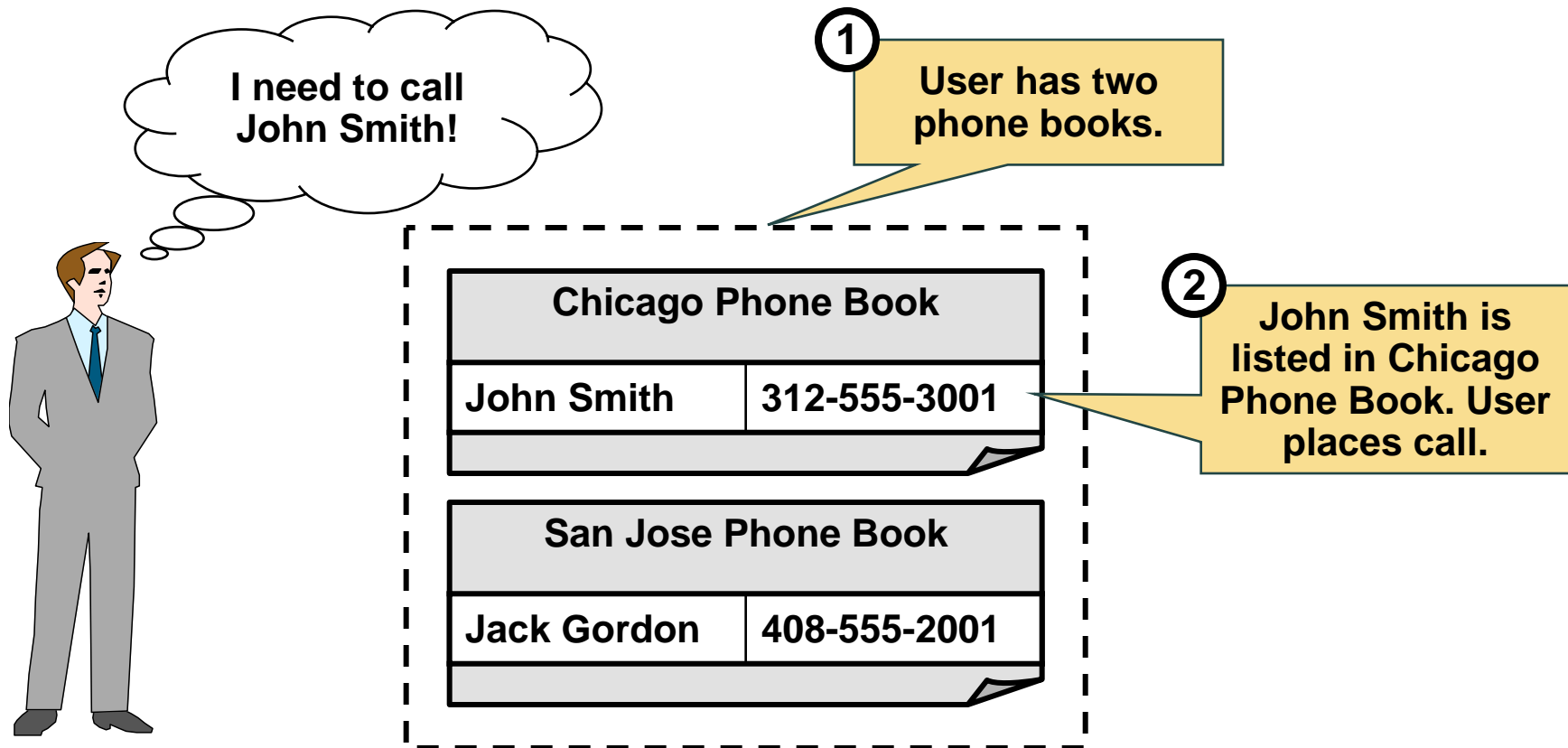


# Partitions and Calling Search Spaces

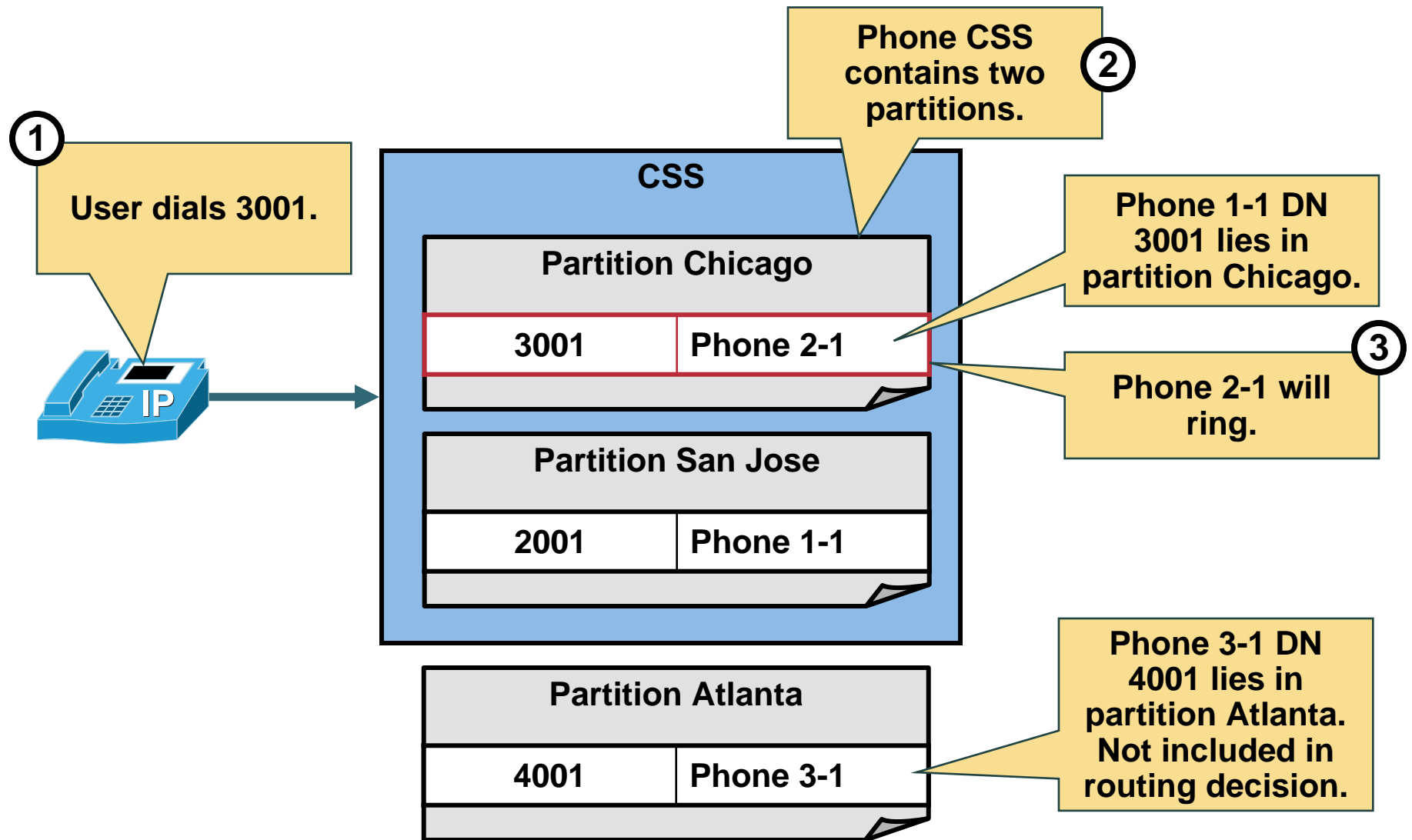
**Partitions and Calling Search Spaces (CSSs) create multiple routing contexts:**

- **Patterns are assigned to partitions, e.g.:**
  - **Directory Numbers**
  - **Route Patterns**
  - **Translation Patterns**
- **CSS includes partitions and can be tied to device and/or line.**
- **Similar to phone books.**
  - **Partition = Phone book your are listed in**
  - **CSS = Phone books you own and use for look ups**
- **Used for call routing and calling privileges**

# Partitions and Calling Search Spaces



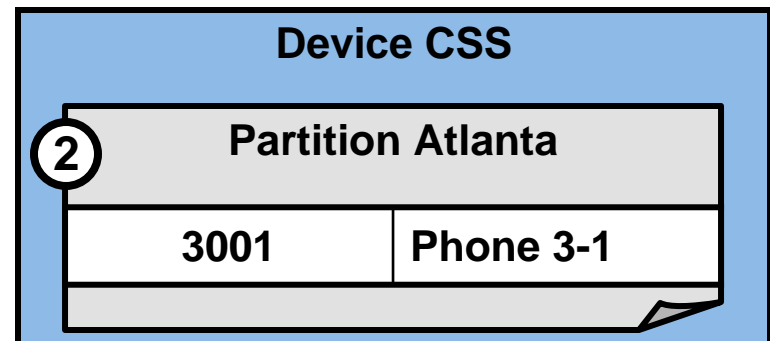
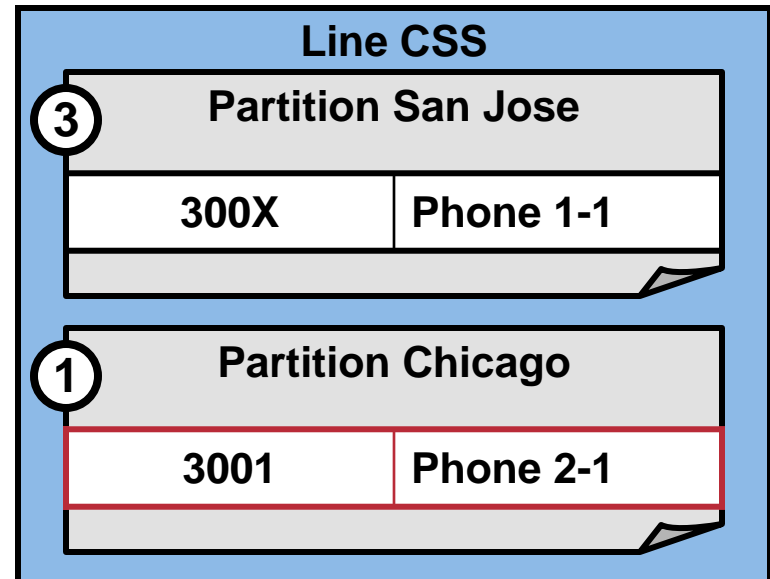
# Partitions and Calling Search Spaces



# Partitions and Calling Search Spaces

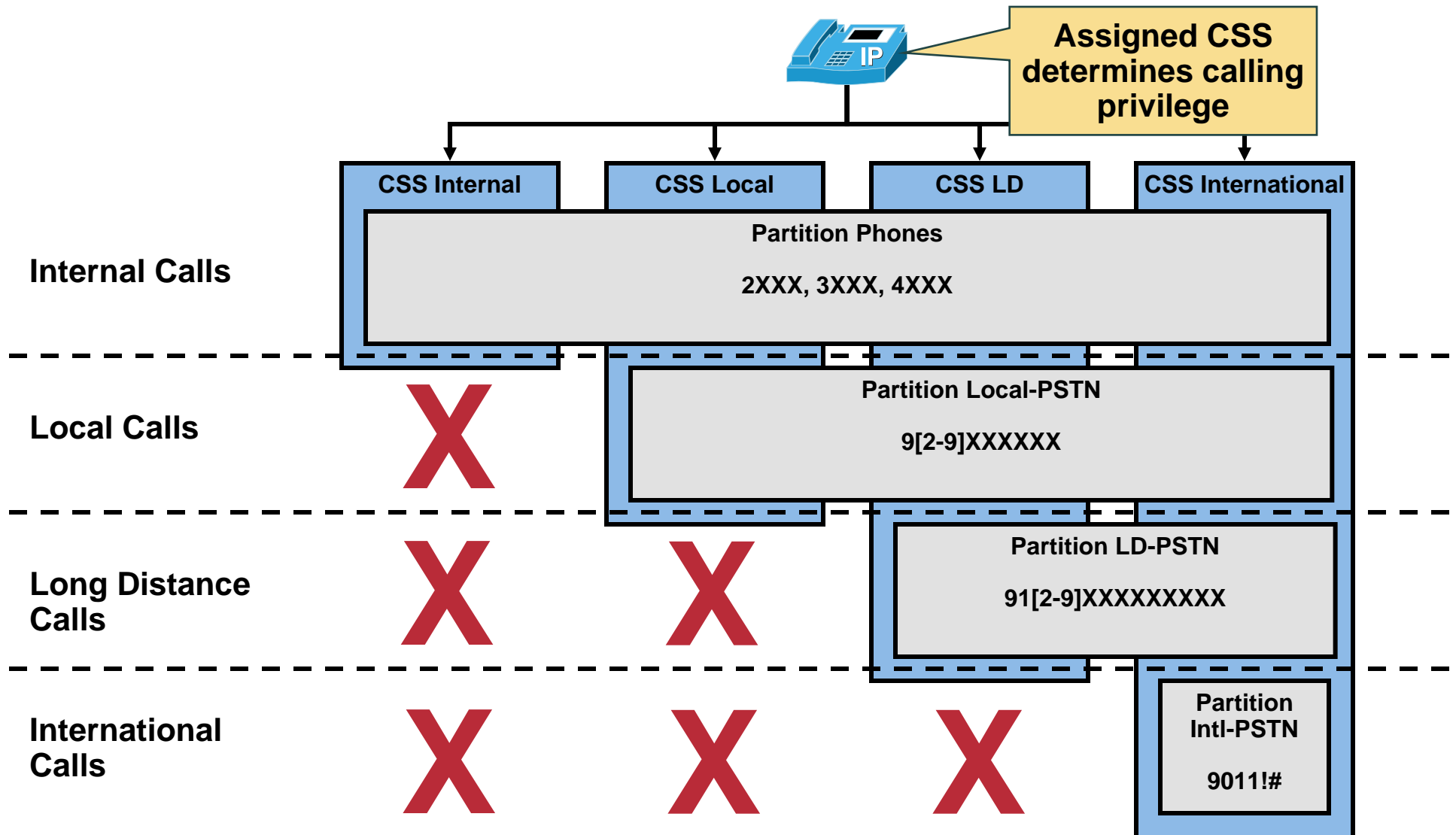
Route selection uses the following rules:

- Closest match routing.
  - Similar to IP routing.
  - X wildcard is less specific.
- Partition order in CSS only used as a tie breaker.
  - Top = Highest Priority
  - Bottom = Lowest Priority
  - Line CSS has higher priority than phone CSS.

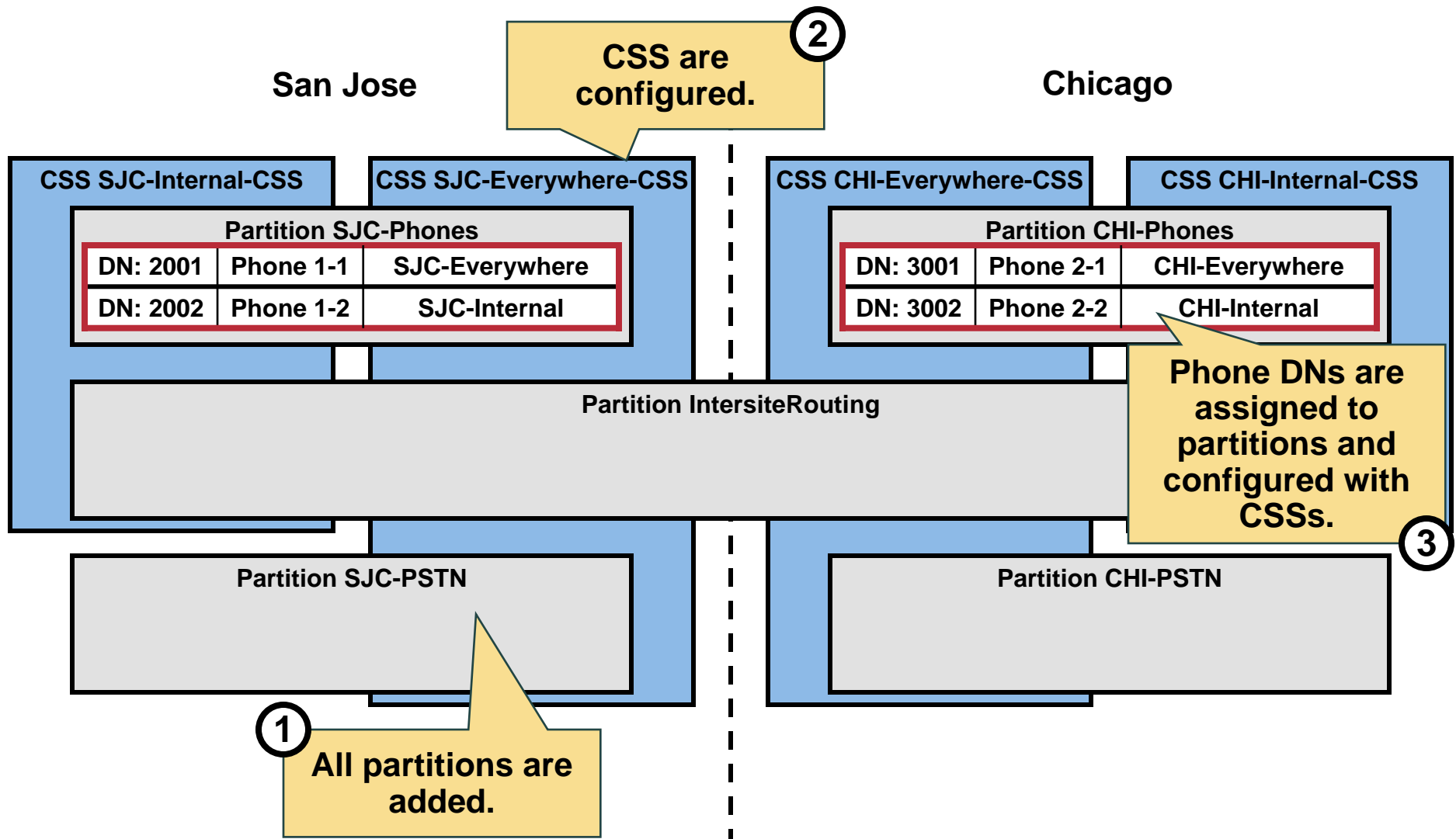




# Partitions and Calling Search Spaces



# Partitions and Calling Search Spaces

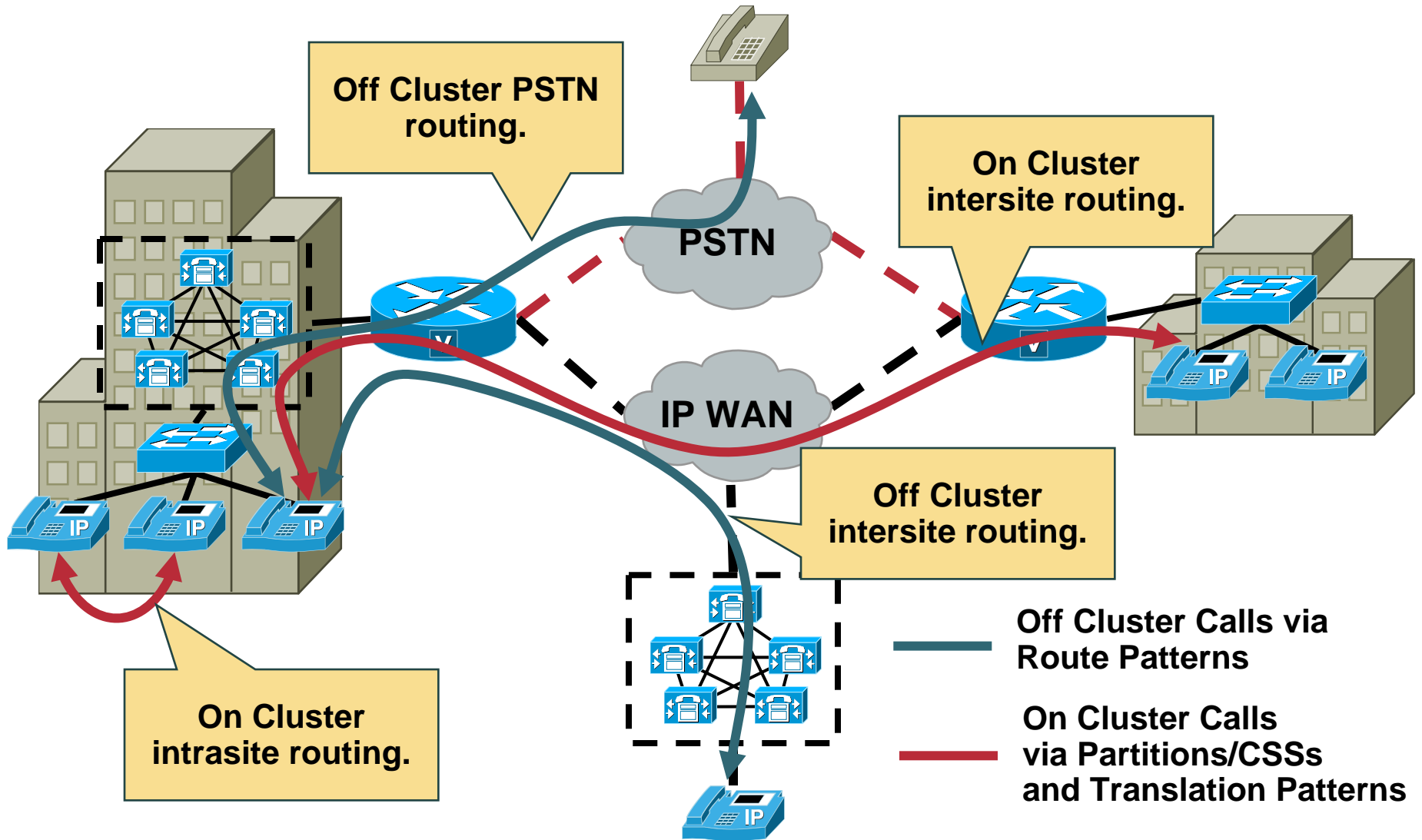


# On/Off Cluster Routing

**CCM differentiates between two types of routing:**

- **On Cluster Routing**
  - **Calls within the same cluster**
  - **Intersite and Intrasite**
  - **Handled by Partitions/CSSs and Translation Patterns**
- **Off Cluster Routing**
  - **Calls destined for off-cluster destinations.**
    - **Other CCM clusters via inter-cluster trunks.**
    - **PSTN or PBXs via voice gateways.**
  - **Handled by Route Patterns.**

# On/Off Cluster Routing

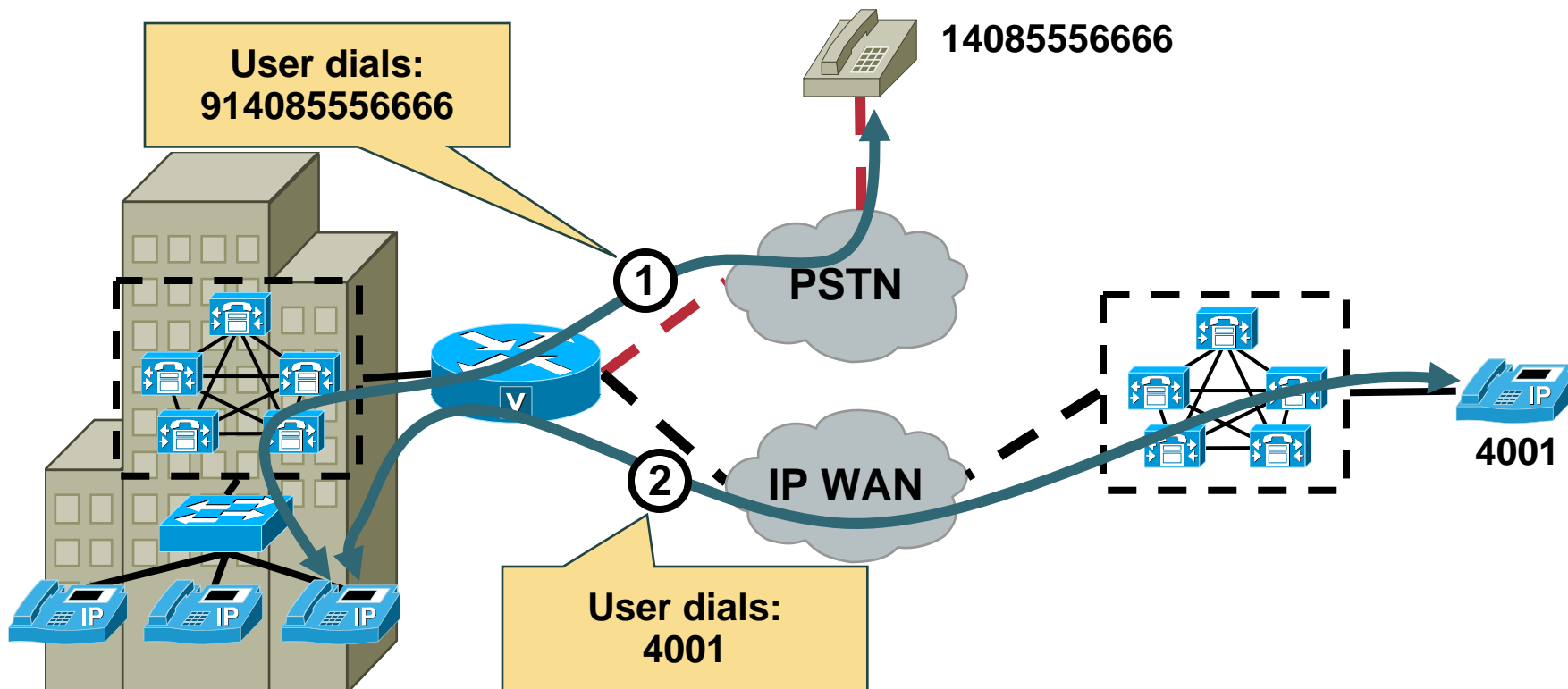


# Route Patterns

**Route Patterns (RPs) are used for Off-Cluster Routing:**

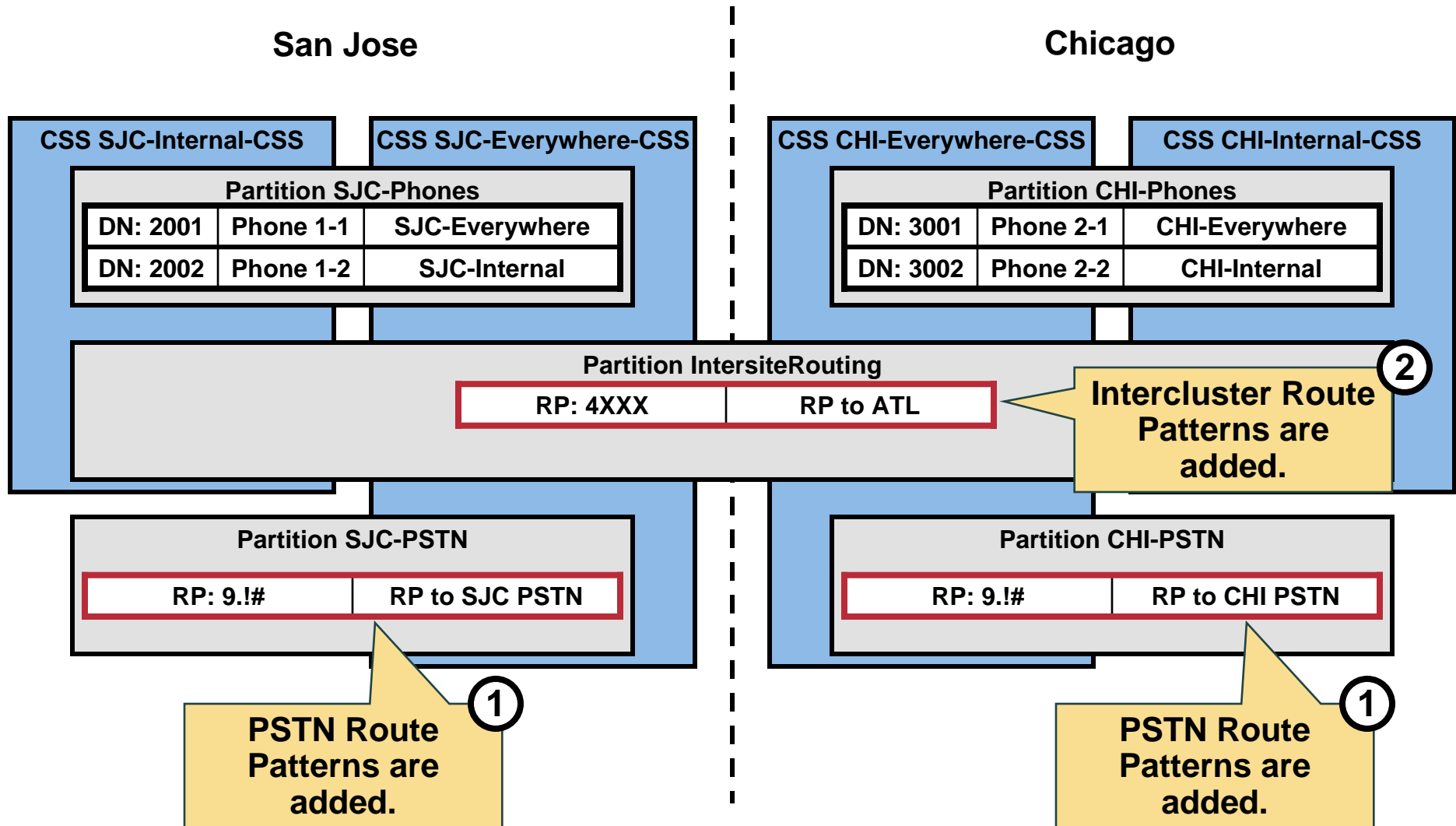
- **Route Pattern assigned to partition.**
- **Destination is either a Route List or Gateway/Trunk**
- **Can also perform digit manipulation:**
  - **Calling Number**
  - **Called Number**

# Route Patterns



	Route Pattern	Partition	Destination	Digit Manipulation
①	9.!#	SJC-PSTN	Router1	Discard 9
②	4XXX	IntersiteRouting	Atlanta CM Trunk	None

# Route Patterns



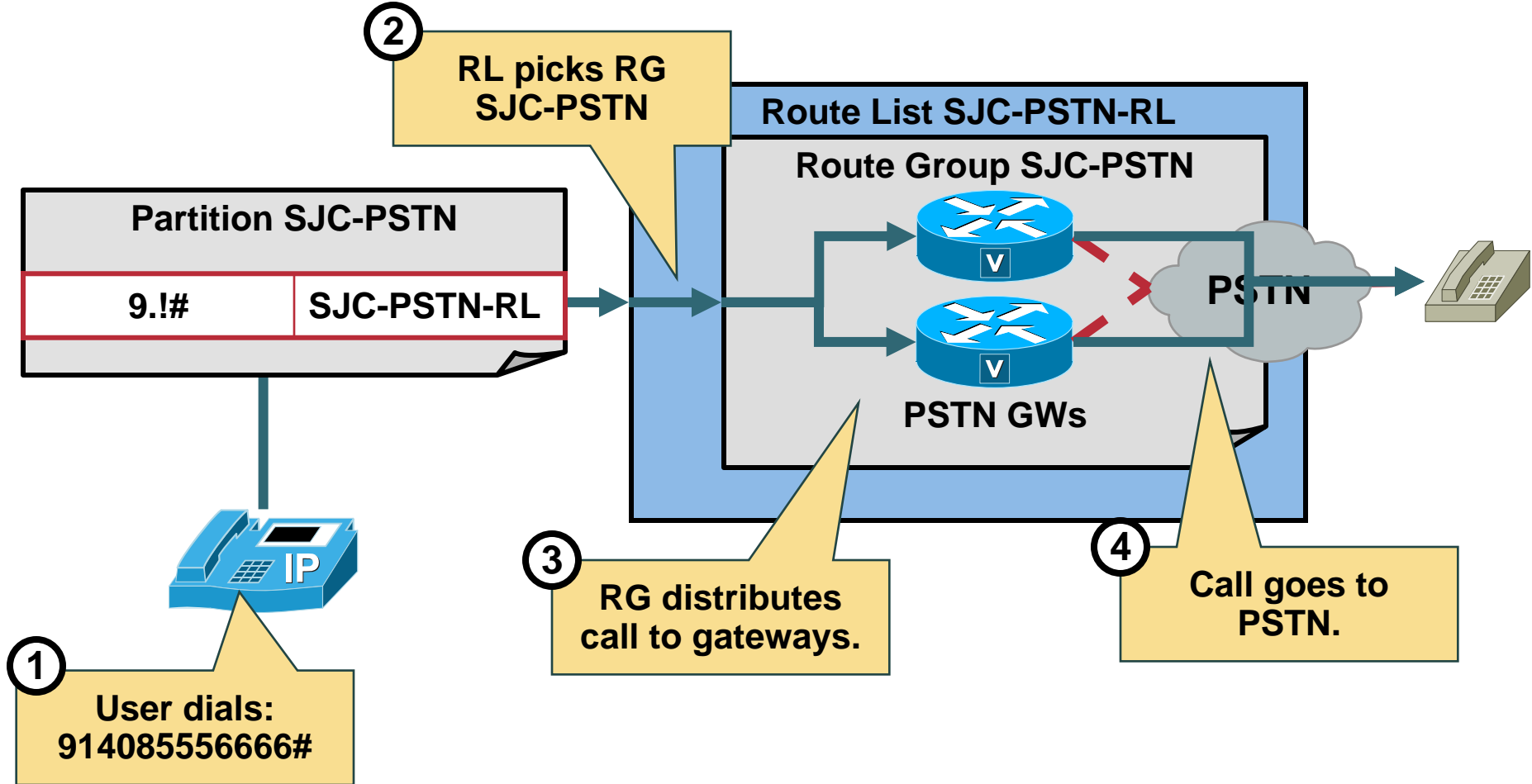
# Route Lists and Route Groups

**Route Lists (RLs) and Route Groups (RGs) group multiple gateways and/or trunks:**

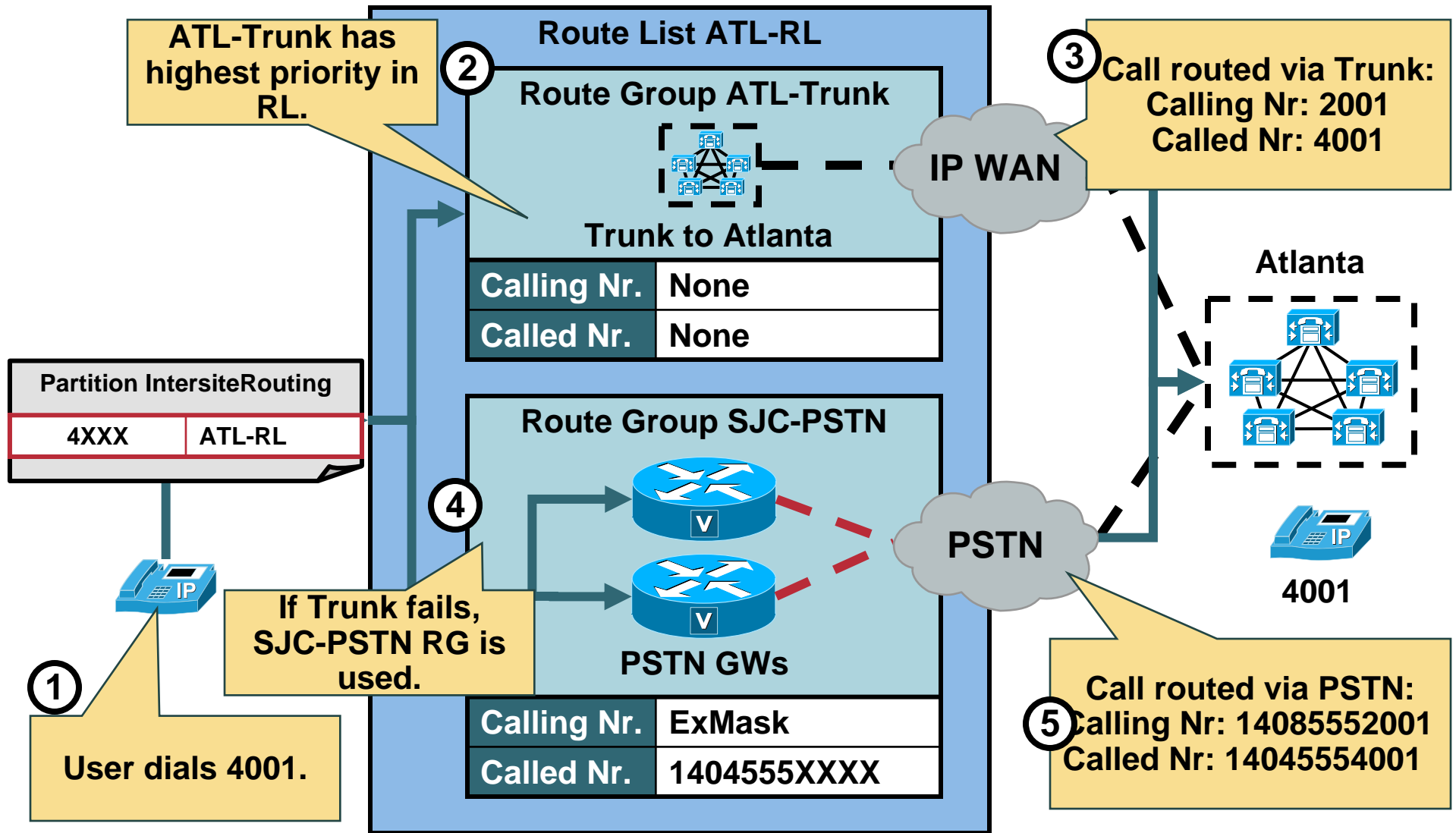
- **Gateways/Trunks assigned to Route Groups.**
- **Route Groups assigned to Route Lists.**
  - **Route Lists can combine multiple Route Groups.**
    - **Top = Highest Priority, Bottom = Lowest Priority**
  - **Each Route Group Entry can have different digit manipulation rules.**
- **Route Patterns point to Route Lists instead of physical gateways.**
  - **Never use physical gateways and/or trunks!**



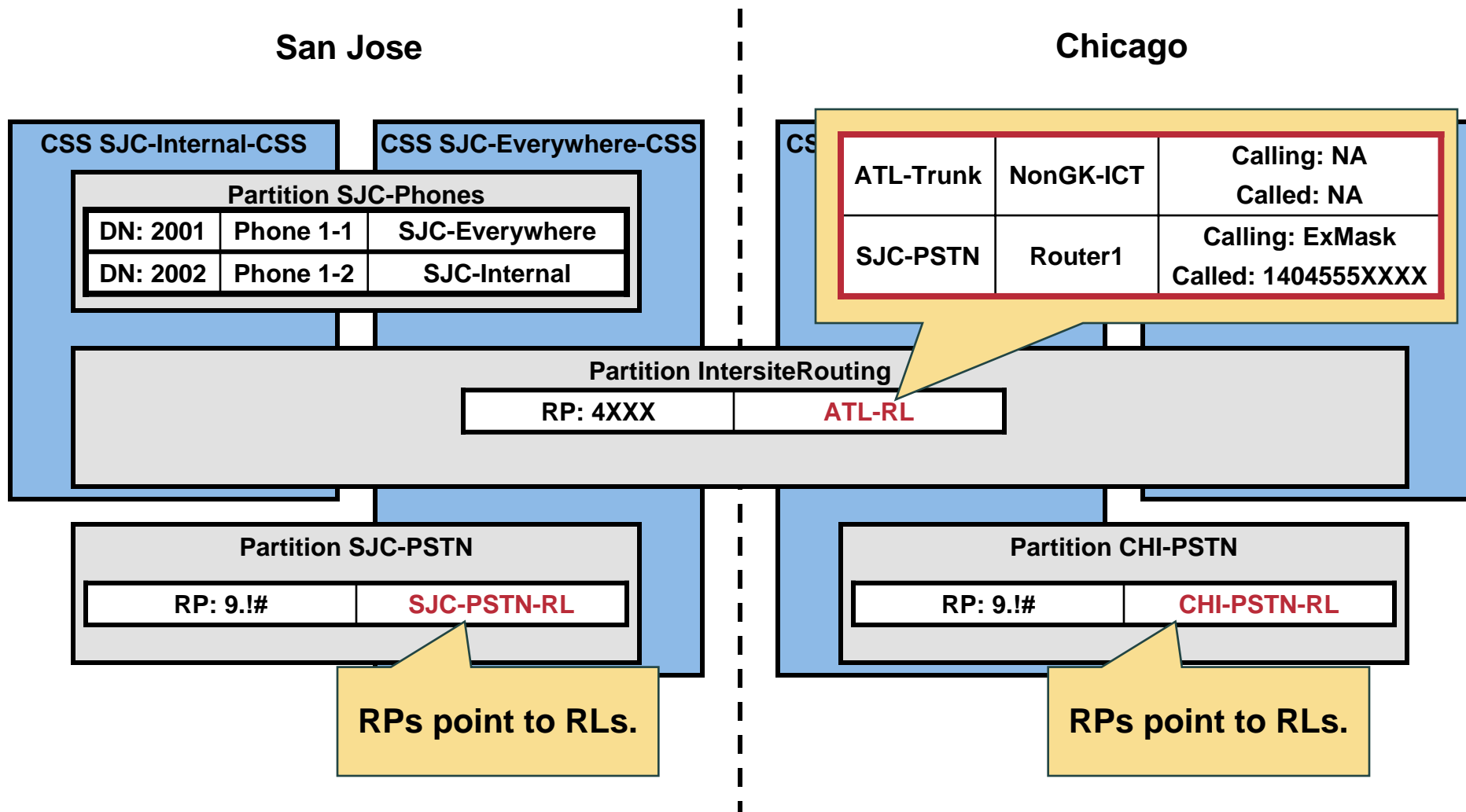
# Route Lists and Route Groups



# Route Lists and Route Groups



# Route Lists and Route Groups

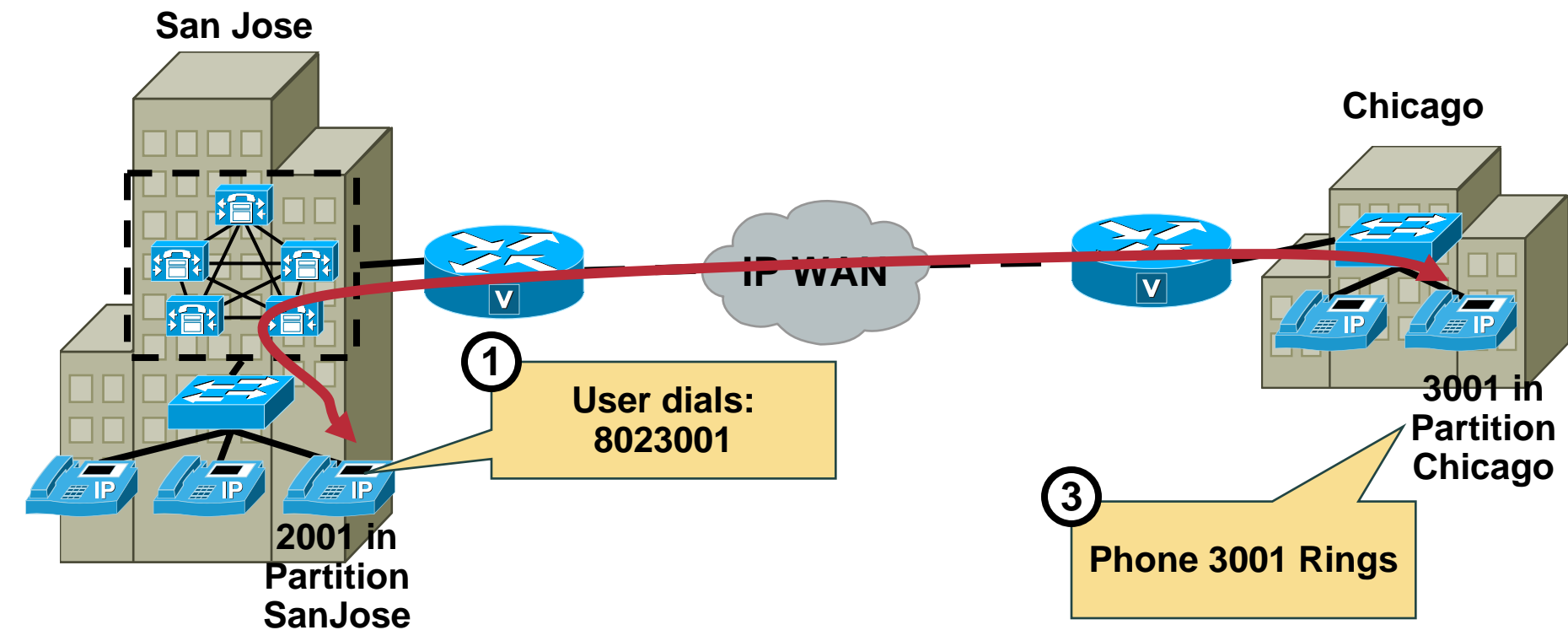


# Translation Patterns

**Translation Patterns are used for on cluster routing:**

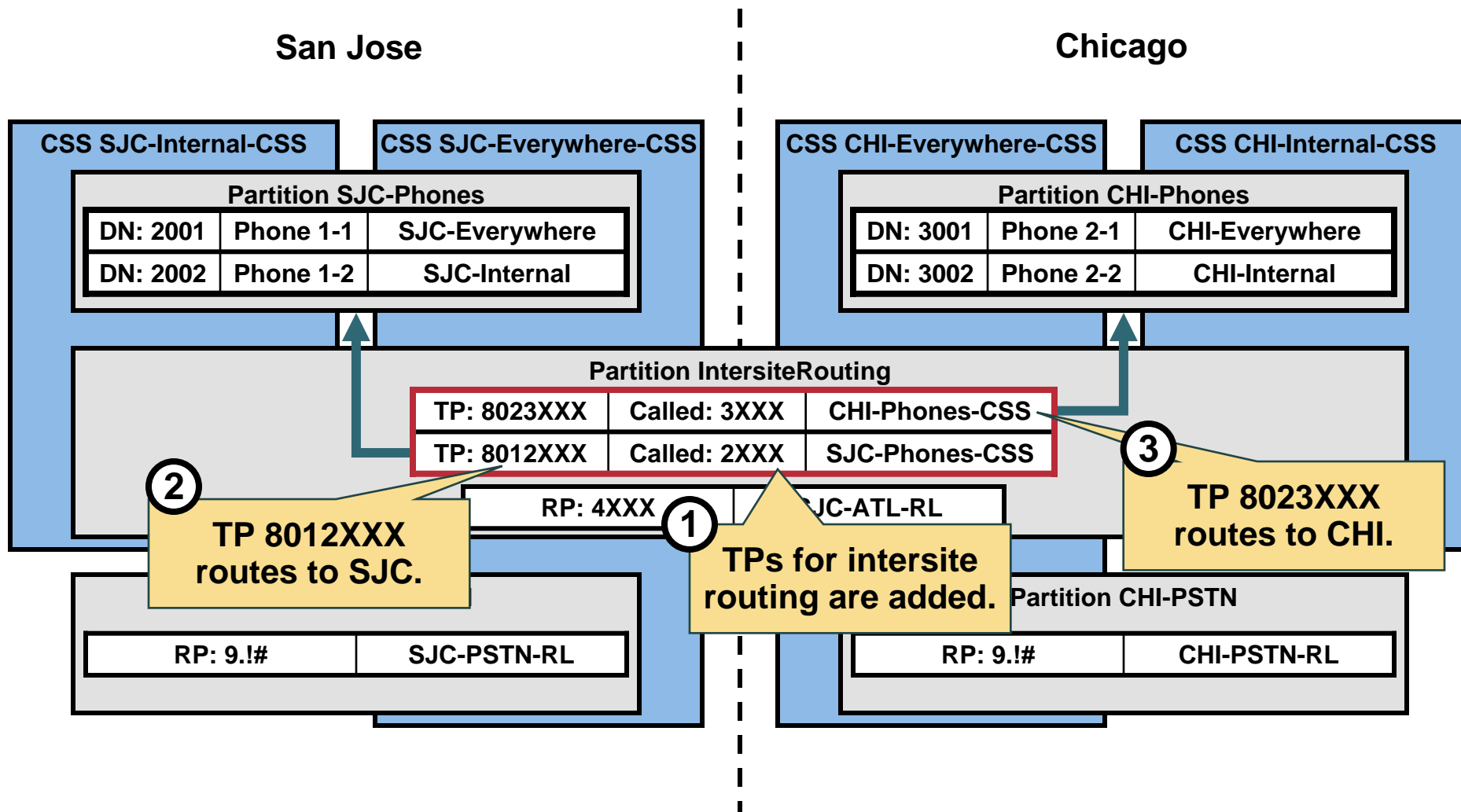
- **Translation Pattern (TPs) assigned to partition.**
- **Destination is a CSS.**
- **Can also perform digit manipulation:**
  - **Calling Number**
  - **Called Number**
- **Often use to solve overlapping DN routing issues.**

# Translation Patterns



Translation Pattern	Partition	Destination	Digit Manipulation
8012XXX	IntersiteRouting	SJC-Phones-CSS	Discard 801
8023XXX	IntersiteRouting	CHI-Phones-CSS	Discard 802

# Translation Patterns

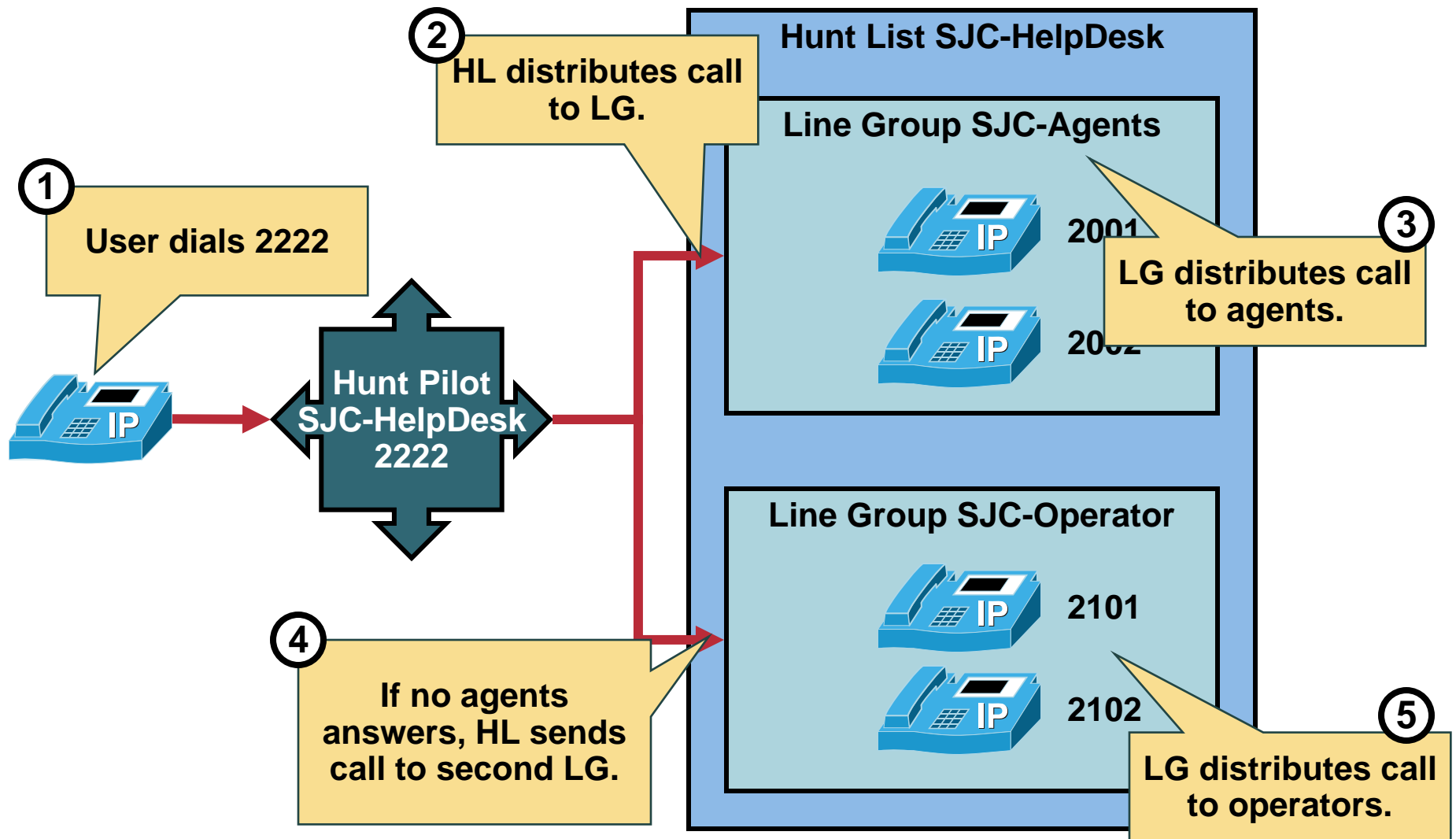


# Line Groups, Hunt Lists and Hunt Pilots

**Line Groups, Hunt Lists and Hunt Pilots are used for call coverage:**

- **Phone DNs assigned to Line Groups (LGs)**
- **Line Groups assigned to Hunt Lists (HLs)**
  - **Can have multiple Line Groups per Hunt List.**
  - **Various hunt algorithms available.**
- **Hunt Lists are assigned to Hunt Pilots**
  - **Hunt Pilot = DN users dial to reach a Hunt List.**

# Line Groups, Hunt Lists and Hunt Pilots





# Review of Cisco Unified CallManager Dial Plan Components

Dial Plan Functionality	CCM Component	T. E.1
Endpoint Addressing	<ul style="list-style-type: none"><li>•Directory Numbers</li><li>•Partitions</li></ul>	
Call Routing and Path Selection	<ul style="list-style-type: none"><li>•Partitions and Calling Search Spaces</li><li>•Route Patterns</li><li>•Route Groups and Route Lists</li><li>•Translation Patterns</li></ul>	
Digit Manipulation	<ul style="list-style-type: none"><li>•Route Patterns</li><li>•Route Groups and Route Lists</li><li>•Translation Patterns</li></ul>	
Call Privileges	<ul style="list-style-type: none"><li>•Partitions</li><li>•Calling Search Spaces</li></ul>	
Call Coverage	<ul style="list-style-type: none"><li>•Line Group, Hunt List and Hunt Pilot</li></ul>	

**Slide 25**

---

**T. E.1**

**Make build slide.**

Tolga Erdogan, 3/15/2006

# Summary

- **Cisco Unified CallManager uses many components to build a dial plan.**
- **Partitions and Calling Search Spaces are used for controlling routing.**
- **Cisco Unified CallManager differentiates between On/Off Cluster Routing**
- **Route Patterns are used for Off Cluster Routing**
- **Route Lists and Route Groups logically bind multiple gateways and trunks together.**
- **Translation Patterns are used for On Cluster Routing and Digit Manipulation.**
- **Hunt Lists and Line Groups are used for call coverage.**

# CISCO SYSTEMS





## Deploying a Dial Plan on CallManager 5.0

# Configuring CallManager Voice Gateways

# Objectives

- **Configuring MGCP Gateways in Cisco Unified CallManager**
- **Configuring Gateways for MGCP**
- **Verify MGCP Gateway Configurations**
- **Configuring H.323 Gateways in Cisco Unified CallManager**
- **Configuring Gateways for H.323**
- **Verify H.323 Gateway Configurations**

# Configuring MGCP Gateways in Cisco Unified CallManager

## CallManager MGCP Gateway configuration steps:

1. Add MGCP Gateway.
2. Add slots.
3. Add VICs.
4. Configure VICs.

# Configuring MGCP Gateways in Cisco Unified CallManager: Add Gateway

The screenshot shows the Cisco CallManager Administration interface in a Mozilla Firefox browser window. The page title is "Find and List Gateway" and the URL is "https://192.168.1.10:8443/ccmadmin/gatewayFindList.do". The navigation bar includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". A menu is open under the "Device" tab, with "Gateway" selected and highlighted in blue. A red box highlights the "Add New" button in the "Search Results" section. Two yellow callout boxes with numbered circles provide instructions: "1 Select Device > Gateway" and "2 Add a new gateway to CallManager."

Find and List Gateway

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Find and List Gateway

Search Options

Find Gateways where Name beg

Search Results

No active query. Please enter your search criteria using the options

Add New

1 Select Device > Gateway

2 Add a new gateway to CallManager.



# Configuring MGCP Gateways in Cisco Unified CallManager: Add Gateway

The screenshot shows the 'Add a new Gateway' page in the Cisco CallManager Administration interface. The page title is 'Add a new Gateway - Mozilla Firefox' and the URL is 'https://192.168.1.10:8443/ccmadmin/gatewayAdd.do'. The page is titled 'Cisco CallManager Administration' and shows the user is logged in as 'CCMAdministrator'. The main heading is 'Add a new Gateway' with a green arrow icon. Below the heading, there is a section titled 'Select the type of gateway you would like to add:'. A dropdown menu is open, showing a list of gateway types. The 'Cisco 2821' option is selected and highlighted in blue. A yellow callout box with a circled '4' points to a 'Next' button, with the text 'Click next.' A second yellow callout box with a circled '3' points to the 'Cisco 2821' option in the dropdown menu, with the text 'Select the appropriate gateway type.' The page also includes a 'Log Off' button and 'Related Links: Back To Find/List'.

4 Click next.

3 Select the appropriate gateway type.

# Configuring MGCP Gateways in Cisco Unified CallManager: Add Gateway

The screenshot shows the 'Add a new Gateway' page in the Cisco CallManager Administration interface. The page title is 'Add a new Gateway - Mozilla Firefox' and the URL is 'https://192.168.1.10:8443/ccmadmin/gatewayAdd.do?product=30040'. The page is titled 'Cisco CallManager Administration' and shows the user is logged in as 'CCMAdministrator'. The main heading is 'Add a new Gateway' with a green arrow icon. Below this, there is a section titled 'Select the type of gateway you would like to add:'. The 'Gateway Type' is set to 'Cisco 2821'. The 'Protocol \*' dropdown menu is open, showing options: 'MGCP', '- Not Selected -', 'MGCP', and 'SCCP'. The 'MGCP' option is highlighted. A yellow callout box with the number '5' points to the 'MGCP' option with the text 'Select the MGCP as Protocol.'. Another yellow callout box with the number '6' points to a 'Next' button with the text 'Click next.'. The 'Next' button is highlighted with a red box. The page also includes a 'Log Off' button and 'Related Links: Back To Find/List'.

# Configuring MGCP Gateways in Cisco Unified CallManager: Add Slot

The screenshot shows the 'Gateway Configuration' page in the Cisco CallManager Administration interface. The page is divided into several sections: Gateway Details, Configured Slots, VICs and Endpoints, and Product Specific Configuration. Four callouts are present:

- 1** (top right): Specify Domain Name (i.e. router host name + ip domain-name), Description and CallManager Group. This callout points to the 'Domain Name \*', 'Description', and 'Cisco CallManager Group' fields.
- 2** (middle right): Select the appropriate slot and module. This callout points to the 'Module in Slot 0' dropdown menu.
- 3** (bottom right): Select the appropriate ISDN switch type. (optional). This callout points to the 'Global ISDN Switch Type' dropdown menu.
- 4** (left side): Click Save. This callout points to the 'Save' button at the bottom left of the page.

The 'Gateway Details' section shows the following configuration:

Product	Cisco 2821
Protocol	MGCP
Domain Name *	Router1
Description	Router1
Cisco CallManager Group	Default

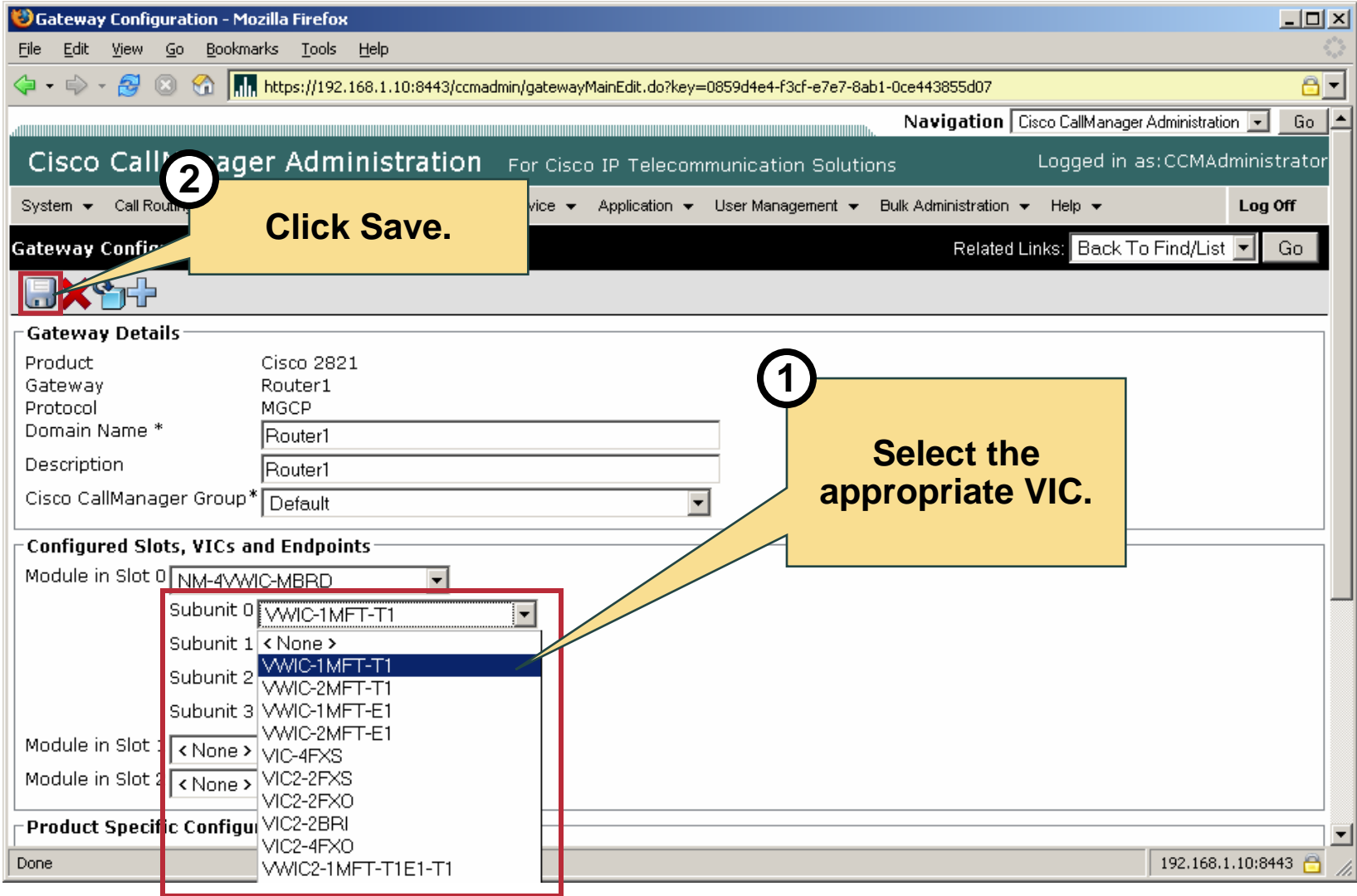
The 'Configured Slots, VICs and Endpoints' section shows the following configuration:

Module in Slot 0	NM-4VVIC-MBRD
Module in Slot 1	< None >
Module in Slot 2	< None >

The 'Product Specific Configuration' section shows the following configuration:

Global ISDN Switch Type	4ESS
Switchback Timing *	Graceful
Switchback uptime-delay (min)	10
Switchback schedule (hh:mm)	12:00

# Configuring MGCP Gateways in Cisco Unified CallManager: Add VICs



The screenshot shows the Cisco CallManager Administration interface for configuring a gateway. The page title is "Gateway Configuration - Mozilla Firefox" and the URL is "https://192.168.1.10:8443/ccmadmin/gatewayMainEdit.do?key=0859d4e4-f3cf-e7e7-8ab1-0ce443855d07". The user is logged in as "CCMAdministrator".

**Callout 1:** A yellow callout box with a circled "1" points to the "Subunit 2" dropdown menu in the "Configured Slots, VICs and Endpoints" section. The dropdown is open, showing a list of options: "VVIC-1MFT-T1", "< None >", "VVIC-1MFT-T1", "VVIC-2MFT-T1", "VVIC-1MFT-E1", "VVIC-2MFT-E1", "VIC-4FXS", "VIC2-2FXS", "VIC2-2FXO", "VIC2-2BRI", "VIC2-4FXO", and "VVIC2-1MFT-T1E1-T1". The "VVIC-1MFT-T1" option is highlighted in blue.

**Callout 2:** A yellow callout box with a circled "2" points to the "Save" button in the top left corner of the configuration area.

**Gateway Details:**

- Product: Cisco 2821
- Gateway: Router1
- Protocol: MGCP
- Domain Name \*: Router1
- Description: Router1
- Cisco CallManager Group\*: Default

**Configured Slots, VICs and Endpoints:**

- Module in Slot 0: NM-4VVIC-MBRD
- Subunit 0: VVIC-1MFT-T1
- Subunit 1: < None >
- Subunit 2: VVIC-1MFT-T1
- Subunit 3: VVIC-2MFT-T1
- Module in Slot 1: < None >
- Module in Slot 2: < None >

**Product Specific Config:**

- VVIC-2BRI
- VVIC-4FXO
- VVIC2-1MFT-T1E1-T1


# Configuring MGCP Gateways in Cisco Unified CallManager: Configure VICs

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The page title is "Gateway Configuration - Mozilla Firefox" and the URL is "https://192.168.1.10:8443/ccmadmin/gatewayMainEdit.do?key=0859d4e4-f3cf-e7e7-8ab1-0ce443855d07". The page is titled "Gateway Configuration" and shows the following details:

**Gateway Details**

- Product: Cisco 2821
- Gateway: Router1
- Protocol: MGCP
- Domain Name \*: Router1
- Description: Router1
- Cisco CallManager Group\*: Default

**Configured Slots, VICs and Endpoints**

Module in Slot	Subunit	Configuration Icon
Module in Slot 0: NM-4VVIC-MBRD	Subunit 0: VWIC-1MFT-T1	0/0/0 
Module in Slot 0	Subunit 1: < None >	
Module in Slot 0	Subunit 2: < None >	
Module in Slot 0	Subunit 3: < None >	
Module in Slot 1: < None >		
Module in Slot 2: < None >		

**Product Specific Configuration**

A callout box with a circled "1" points to the configuration icon next to the "0/0/0" entry in the Subunit 0 row, with the text "Click configuration icon to configure VIC".

# Configuring MGCP Gateways in Cisco Unified CallManager: Configure VICs

The screenshot shows the Cisco CallManager Administration interface in a Mozilla Firefox browser window. The page title is "Gateway Configuration" and the URL is "https://192.168.1.10:8443/ccmadmin/gatewayEdit.do?product=52&slot=0&subunit=0&port=0&endpoint=50/SU0/DS1-0&mgcpid=0859d4e4-f3cf-e7e7". The page is titled "Gateway Configuration" and includes a navigation menu with options like "System", "Call Routing", "Device", "Application", "User Management", "Bulk Administration", and "Help". A "Log Off" button is visible in the top right. The main content area is titled "Select Protocol for this Gateway" and contains a "Device Protocol" dropdown menu. The dropdown menu is open, showing options: "Digital Access PRI", "- Not Selected -", "Digital Access T1", and "Digital Access PRI". A "Next" button is located below the dropdown. A "Done" button is at the bottom left of the page. The status bar at the bottom right shows the IP address "192.168.1.10:8443".

**3** Click Next.

**2** Select the appropriate device protocol.

# Configuring MGCP Gateways in Cisco Unified CallManager: Configure VICs

The screenshot shows the 'Gateway Configuration' web interface in Mozilla Firefox. The interface is divided into several sections, each with a corresponding callout box on the right:

- Device Information (Generic):** This section includes fields for Product (Cisco MGCP T1 Port), Gateway (Router1), Device Protocol (Digital Access PRI), Registration (Registered with Cisco CallManager 192.168.1.10), IP Address (192.168.1.254), End-Point Name (S1/DS1-0@Router1), Description (S1/DS1-0@Router1), Device Pool (SanJose), Call Classification (Use System Default), NetworkLocale (< None >), Media Resource Group List (< None >), Location (SJC), and AAR Group (< None >). There are also checkboxes for 'Transmit UTF-8 for Calling Party Name' and 'V150 (subset)'.
- MLPP Information (Generic):** This section includes fields for MLPP Domain (< None >), MLPP Indication (Default), and MLPP Preemption (Default).
- Interface Information (VIC Specific):** This section includes fields for PRI Protocol Type (PRI NI2), Protocol Side (User), Channel Selection Order (Bottom Up), Channel IE Type (Use Number when 1B), PCM Type (µ-law), Delay for first restart (32), and Delay between restarts (4). There are also checkboxes for 'Inhibit restarts at PRI initialization', 'Enable status poll', and 'Unattended Port'.
- Call Routing Information - Inbound Calls (Generic):** This section includes fields for Significant Digits (All), Calling Search Space (SJC-Phones-CSS), AAR Calling Search Space (< None >), and Prefix DN.
- Call Routing Information - Outbound Calls (Generic):** This section includes fields for Calling Party Presentation (Default) and Calling Party Selection.

# Configuring MGCP Gateways in Cisco Unified CallManager: Configure VICs

**Call Routing Information - Outbound Calls (Generic)**

- Calling Party Presentation\*: Default
- Calling Party Selection\*: Originator
- Called party IE number type unknown\*: Cisco CallManager
- Calling party IE number type unknown\*: Cisco CallManager
- Called Numbering Plan\*: Cisco CallManager
- Calling Numbering Plan\*: Cisco CallManager
- Number of digits to strip\*: 0
- Caller ID DN
- SMDI Base Port\*

**PRI Protocol Type Specific Information. (VIC Specific)**

- Display IE Delivery
- Redirecting Number IE Delivery - Outbound
- Redirecting Number IE Delivery - Inbound
- Send Extra Leading Character in Display IE\*\*\*
- Setup non-ISDN Progress Indicator IE Enable\*\*\*\*
- MCDN Channel Number Extension Bit Set to Zero\*\*
- Send Calling Name In Facility IE
- Interface Identifier Present\*\*
- Interface Identifier Value\*\*
- Connected Line ID Presentation (QSIG Inbound Call)\*: Default

**UUIE Configuration. (VIC Specific)**

- Passing Precedence Level Through UUIE
- Security Access Level\* 2

**Product Specific Configuration. (VIC Specific)**

- Line Coding\*: B8ZS
- Framing\*: ESF
- Clock\*: External
- Input Gain (-6..14 db)\*: 0
- Output Attenuation (-6..14 db)\*: 0
- Echo Cancellation Enable\*: Enable
- Echo Cancel Coverage (ms)\*: 8

**Save/Reset Configuration.**

Save Delete Reset



# Configuring MGCP Gateways in Cisco Unified CallManager: Configure VICs

## Most important MGCP gateway configuration settings:

	Description	Example
Device Information	Includes device pool and location configuration.	Device Pool: SanJose Location: SanJose
Interface Information	Contains all settings used to configure the VIC, e.g. ISDN Layer 3	Switch Type: PRI NI2 Protocol Side: User
Call Routing Inbound Calls	Digit manipulation and routing for inbound calls.	Significant Digits: 4 Inbound CSS: None
Call Routing Outbound Calls	Digit manipulation and routing for outbound calls.	Number of Digits to strip: 0
VIC Specific Information	VIC dependant configuration, e.g. ISDN Layer 2	Line Code: B8ZS Framing: ESF

# Configuring Gateways for MGCP

## **IOS MGCP gateway configuration steps:**

- 1. Specify configuration server.**
- 2. Enable configuration download.**

# Configuring Gateways for MGCP

```
router(config)#
```

```
ccm-manager config server <CCM TFTP IP>
```

- **Specifies CallManager TFTP server hosting the gateway config XML file.**

```
router(config)#
```

```
ccm-manager config
```

- **Activates gateway to pull configuration from TFTP server.**

```
router#
```

```
show ccm-manager
```

- **Verify connection to CallManager**

# Configuring Gateways for MGCP

router#

```
show mgcp endpoint
```

- Displays mgcp controlled interfaces and channels.

router#

```
show isdn status
```

- Displays ISDN Layer 1, Layer 2 and Layer 3 information

router#

```
debug isdn q931
```

- Displays ISDN Q931 information.

# Verifying MGCP Gateway Operation

```
Router1(config)#ccm-manager config server 192.168.1.10
```

```
Router1(config)#ccm-manager config
```

```
Router1#show ccm-manager
```

```
MGCP Domain Name: Router1
```

```
Priority          Status          Host
```

```
=====
```

```
Primary          Registered      192.168.1.10
```

```
First Backup     None
```

```
Second Backup    None
```

```
...
```

```
Router1#sho mgcp endpoint
```

Registered with  
CallManager.

```
Interface T1 1/0
```

ENDPOINT-NAME	V-PORT	SIG-TYPE	ADMIN
S1/ds1-0/1@Router1	1/0:23	none	up
S1/ds1-0/2@Router1	1/0:23	none	up
S1/ds1-0/3@Router1	1/0:23	none	up
S1/ds1-0/4@Router1	1/0:23	none	up
S1/ds1-0/5@Router1	1/0:23	none	up
S1/ds1-0/6@Router1	1/0:23	none	up
S1/ds1-0/7@Router1	1/0:23	none	up
S1/ds1-0/8@Router1	1/0:23	none	up

Registered MGCP  
endpoints.

```
...
```

# Configuring Gateways for MGCP

```
Router1#sho isdn status
Global ISDN Switchtype = primary-ni

%Q.931 is backhauled to CCM MANAGER 0x0003 on DSL 0. Layer 3 output may not
appl
y

ISDN Serial1/0:23 interface
    dsl 0, interface ISDN Switchtype = primary-ni
    L2 Protocol = Q.921 0x0000 L3 Protocol(s) = CCM MANAGER 0x0003
Layer 1 Status:
    ACTIVE
Layer 2 Status:
    TEI = 0, Ces = 1, SAPI = 0, State = MULTIPLE_FRAME_ESTABLISHED
Layer 3 Status:
    0 Active Layer 3 Call(s)
Active dsl 0 CCBS = 0
The Free Channel Mask: 0x807FFFFFFF
Number of L2 Discards = 0, L2 Session ID = 2
Total Allocated ISDN CCBS = 0
...
```

Backhauled to  
CallManager.

Layer 2 up.

# Configuring H.323 Gateways in Cisco Unified CallManager

## CallManager H.323 Gateway configuration steps:

1. Add H.323 Gateway.
2. Configure H.323 Gateway settings.

# Configuring H.323 Gateways in Cisco Unified CallManager: Add Gateway

The screenshot shows the Cisco CallManager Administration interface in a Mozilla Firefox browser window. The page title is "Find and List Gateway" and the URL is "https://192.168.1.10:8443/ccmadmin/gatewayFindList.do". The interface includes a navigation bar with "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". A menu bar contains "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Device" menu is open, showing options: "CTI Route Point", "Gatekeeper", "Gateway", "Phone", "Trunk", and "Device Settings". The "Gateway" option is highlighted with a red box and a callout bubble labeled "1" containing the text "Select Device > Gateway". Below the menu, the "Search Options" section has a dropdown menu set to "Name" and a search box containing "beg". The "Add New" button is highlighted with a red box and a callout bubble labeled "2" containing the text "Add a new gateway to CallManager.". The "Search Results" section displays the message "No active query. Please enter your search criteria using the options".



# Configuring H.323 Gateways in Cisco Unified CallManager: Add Gateway

The screenshot displays the 'Add a new Gateway' page in the Cisco CallManager Administration interface. The page title is 'Add a new Gateway - Mozilla Firefox' and the URL is 'https://192.168.1.10:8443/ccmadmin/gatewayAdd.do'. The page is logged in as 'CCMAdministrator'. The main heading is 'Add a new Gateway'. A yellow callout box with a circled '4' and an arrow points to a green 'Next' button with the text 'Click Next.'. Below the heading, there is a section titled 'Select the type of gateway you would like to add:'. A dropdown menu for 'Gateway Type \*' is open, showing a list of gateway models. The 'H.323 Gateway' option is selected and highlighted with a red box. A yellow callout box with a circled '3' points to this option with the text 'Select H.323 Gateway.'. The list of gateway models includes: Cisco 362X, Cisco 364X, Cisco 366X, Cisco 3725, Cisco 3745, Cisco 3825, Cisco 3845, Cisco 72XX, Cisco Catalyst 4000 Access Gateway Module, Cisco Catalyst 4224 Voice Gateway Switch, Cisco Catalyst 6000 24 port FXS Gateway, Cisco Catalyst 6000 E1 VoIP Gateway, Cisco Catalyst 6000 T1 VoIP Gateway, Cisco DE-30+ Gateway, Cisco DT-24+ Gateway, Cisco VG200, Cisco VG248 Gateway, Communication Media Module, H.323 Gateway, and VG224. The status bar at the bottom shows 'Done' and the IP address '192.168.1.10:8443'.

# Configuring H.323 Gateways in Cisco Unified CallManager: Gateway

The screenshot shows the Cisco CallManager Administration interface for configuring an H.323 Gateway. The browser window is titled "Gateway Configuration - Mozilla Firefox" and the URL is "https://192.168.1.10:8443/ccmadmin/gatewayAdd.do?product=17". A yellow callout box with a circled "3" points to the "Save" button, with the text "Click Save.". A modal dialog box is open, displaying a warning icon and the message "Reset the gateway to have the changes take effect." with an "OK" button. The "Device Information" section is highlighted with a red box and contains the following fields:

Product	H.323 Gateway
Device Protocol	H.225
Device Name*	192.168.2.254
Description	Router2
Device Pool*	SanJose
Call Classification*	Use System Default
Media Resource Group List	< None >
Packet Capture Mode*	None
Packet Capture Duration	0
Location*	Hub_None
AAR Group	< None >
Tunneled Protocol*	None
Signaling Port*	1720

Two additional yellow callout boxes provide instructions: a circled "1" points to the "Device Name" field with the text "Device Name = IP Address or resolvable hostname.", and a circled "2" points to the "Description" field with the text "Configure all other required gateway settings.".

# Configuring H.323 Gateways in Cisco Unified CallManager: Gateway

## Most important H.323 gateway configuration settings:

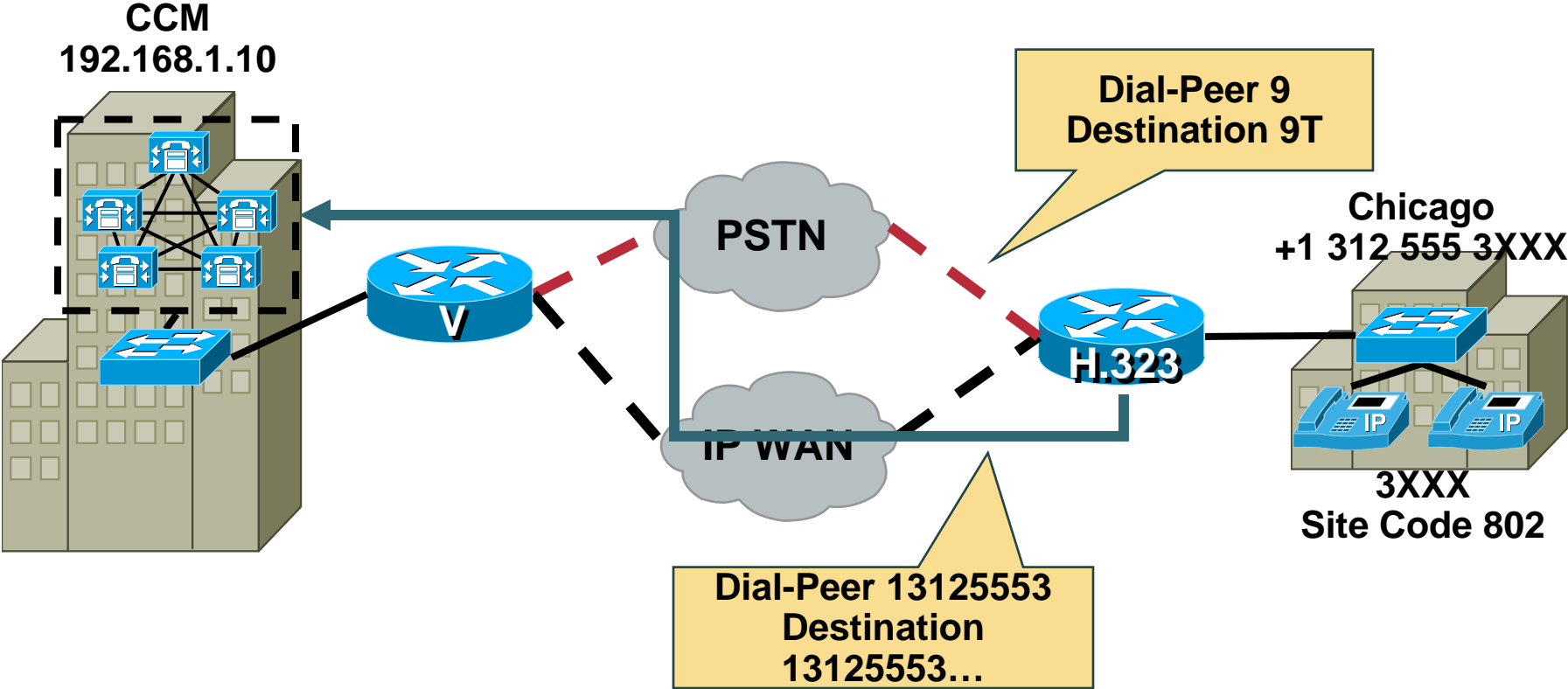
	Description	Example
Device Information	Includes device name, device pool and location configuration.	Device Name: 192.168.2.254 Device Pool: Chicago Location: Chicago
Call Routing Inbound Calls	Digit manipulation and routing for inbound calls.	Significant Digits: 4 Inbound CSS: None
Call Routing Outbound Calls	Digit manipulation and routing for outbound calls.	Number of Digits to strip: 0

# Configuring Gateways for H.323

## **IOS H.323 gateway configuration steps:**

- 1. Verify voice port configuration.**
- 2. Configure H.323 source interface.**
- 3. Configure H.323 dial-peers.**

# Configuring Gateways for H.323



# Configuring Gateways for H.323: Verify Voice Port Configuration

```
Router2#show running-config
...
!
controller T1 1/0
  framing esf
  linecode b8zs
  pri-group timeslots 1-24
!
...
!
interface Serial1/0:23
  no ip address
  encapsulation hdlc
  isdn switch-type primary-ni
  isdn incoming-voice voice
  no cdp enable
!
```

# Configuring Gateways for H.323: Source Interface

```
router(config-if)#
```

```
h323-gateway voip bind srcaddr <interface IP>
```

- Specifies the source IP address used for H.323.

```
Router2#show running-config
...
!
interface FastEthernet0/0
  description Site2 LAN
  ip address 192.168.2.254 255.255.255.0
  duplex auto
  speed auto
  h323-gateway voip bind srcaddr 192.168.2.254
!
```

# Configuring Gateways for H.323: Dial-Peers

```
router(config)#
```

```
dial-peer voice <nr> voip|pots
```

- **Configures a VoIP or POTS dial-peer.**

```
router(config-dial-peer)#
```

```
destination-pattern <pattern>
```

- **Configures the dialed destination of a dial-peer**

```
router(config-dial-peer)#
```

```
session target ipv4:<ip address>
```

- **Configures a H.323 target of a VoIP dial-peer**



# Configuring Gateways for H.323: Dial-Peers

```
router(config-dial-peer)#
```

```
codec <codec type>
```

- **Configures the codec for a VoIP dial-peer.**

```
router(config-dial-peer)#
```

```
port <VIC port>
```

- **Configures a POTS target for POTS dial-peers**

```
router(config-dial-peer)#
```

```
direct-inward-dial
```

- **Enables DID calls.**

```
router(config-dial-peer)#
```

```
incoming called-number
```

- **Specifies the incoming called-number which needs to match.**

# Configuring Gateways for H.323: Dial-Peers

```
Router2#show running-config
!
dial-peer voice 9 pots
 destination-pattern 9T
 incoming called-number 13125552...
 direct-inward-dial
 port 1/0:23
!
dial-peer voice 13125552 voip
 destination-pattern 13125553...
 session target ipv4:192.168.1.10
 incoming called-number 9T
 codec g711ulaw
 no vad
 dtmf-relay h245-alphanumeric
!
```

# Verifying H.323 Gateway Operation

router#

```
show voice port summary
```

- **Displays voice-port information.**

router#

```
debug isdn q931
```

- **Displays ISDN Q931 information.**

# Verifying H.323 Gateway Operation

router#

```
debug voip dialpeer
```

- Debugs dial-peer matching.

```
Router2#debug voip dialpeer
*Apr 10 05:03:40.816: //-1/xxxxxxxxxxxx/DPM/dpAssociateIncomingPeerCore:
  Calling Number=14085556666, Called Number=13125552001, Voice-
Interface=0x855C
AE88,
  Timeout=TRUE, Peer Encap Type=ENCAP_VOICE, Peer Search
Type=PEER_TYPE_VOICE,
  Peer Info Type=DIALPEER_INFO_SPEECH
...
```

# Verifying H.323 Gateway Operation

```
router#
```

```
Debug h225 asn1 | events | q931
```

- Debugs dial-peer matching.

```
Router2#debug h225 asn1
*Apr 10 05:07:38.690: H225 NONSTD OUTGOING PDU ::=
value H323_UU_NonStdInfo ::=
  {
    version 2
    protoParam qsigNonStdInfo :
      {
        iei 4
        rawMesg '04038090A21803A983811E0285836C0D00803134...'H
      }
    progIndParam progIndIEinfo :
      {
        progIndIE '00000003'H
      }
  }
...
```

# Summary

- **MGCP gateway configuration depends on router model and used VICs.**
- **Configure IOS MGCP gateways to pull the configuration from CallManager to reduce manual configuration efforts.**
- **Debug and show commands can be used to verify the MGCP gateway and endpoint status.**
- **H.323 gateway is generic and does not depend on used IOS gateway.**
- **Configure IOS H.323 gateways with correct dial-peers to interact with CallManager.**
- **Debug commands can be used to verify dial-peer matching and H.225 signalling.**

# CISCO SYSTEMS





**Deploying a Dial Plan on CallManager 5.0**

# **Configuring Cisco Unified CallManager Trunks**



# Objectives

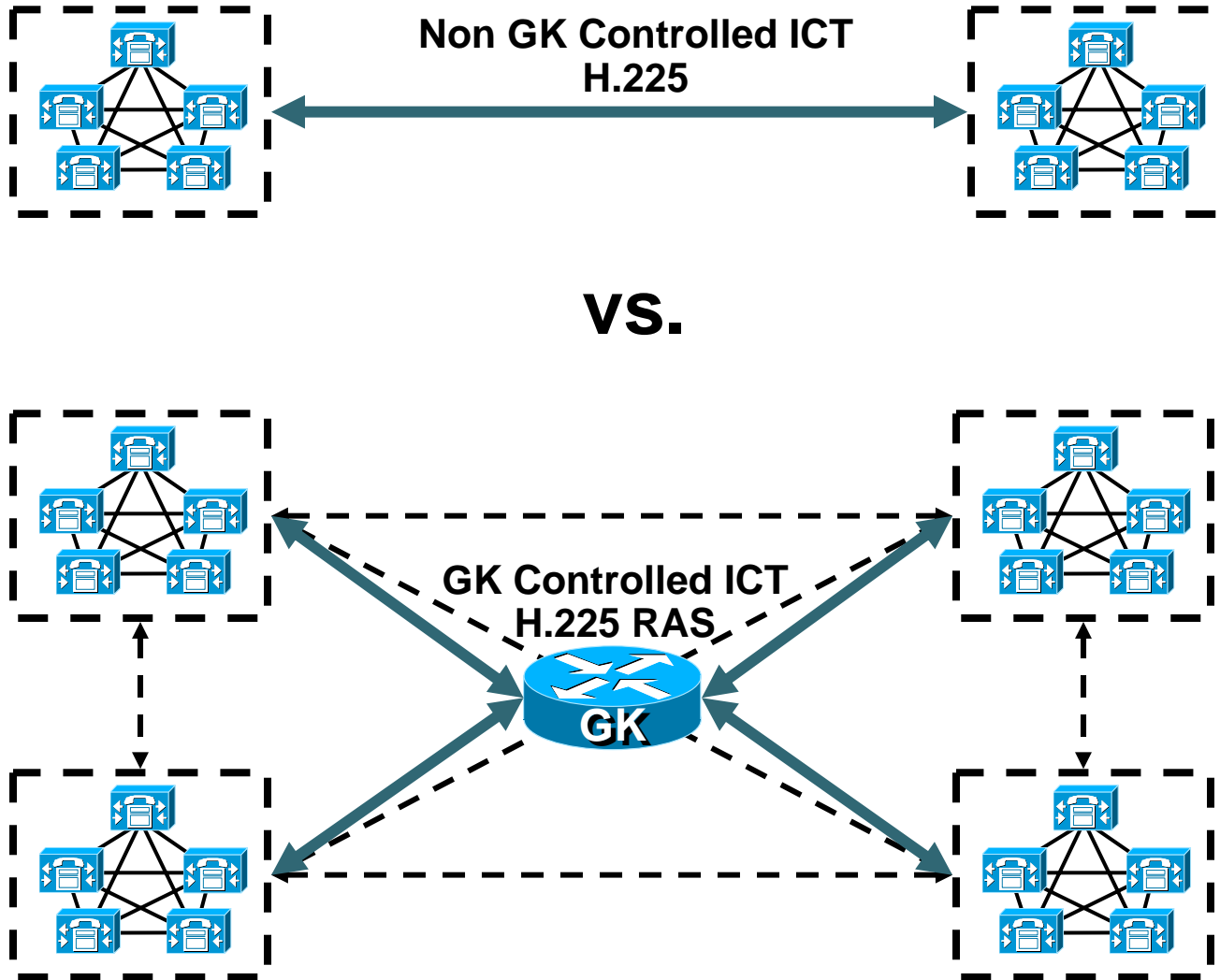
- **Trunk Overview**
- **Configuring Non-Gatekeeper Controlled Intercluster Trunks**
- **Configuring Gatekeeper Controlled Intercluster Trunks**
- **Configuring QSIG for Trunks**
- **Configuring SIP Trunks**

# Trunk Overview

## **Trunks are Off-Cluster VoIP connection:**

- **Either H.323 or SIP**
- **Intercluster Trunks interconnect CCM clusters**
  - **H.323 Non-Gatekeeper controlled**
  - **H.323 Gatekeeper controlled**
- **SIP Trunks used to interwork with other VoIP networks**
  - **SIP Service Providers**
  - **SIP enabled PBXs**
  - **SIP enabled Applications**

# Trunk Overview



# Configuring Non-Gatekeeper Controlled Intercluster Trunks

## **Non-Gatekeeper Controlled Intercluster trunk configuration steps:**

- 1. Add Non-Gatekeeper Controlled intercluster trunk.**
- 2. Specify trunk parameters.**

# Configuring Non-Gatekeeper Controlled Intercluster Trunks: Adding a Trunk

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/trunkFindList.do`. The page title is "Find and List Trunks". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Device" dropdown menu is expanded, showing options: "CTI Route Point", "Gatekeeper", "Gateway", "Phone", "Trunk", and "Device Settings". The "Trunk" option is highlighted with a red box. A yellow callout box with a circled "1" points to this option and contains the text "Select Device > Trunk.". Below the search options, the "Add New" button is highlighted with a red box. A second yellow callout box with a circled "2" points to this button and contains the text "Click Add New.". The search results area shows the message: "No active query. Please enter your search criteria using the options above."

# Configuring Non-Gatekeeper Controlled Intercluster Trunks: Adding a Trunk

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccadmin/trunkEdit.do?prod=77`. The page title is "Trunk Configuration".

At the top, there is a navigation bar with "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". Below this is a menu with "System", "Call Routing", "Device", "Application", "User Management", "Bulk Administration", and "Help". A "Log Off" button is also present.

The main content area shows the "Trunk Configuration" form. A green arrow points to the "Next" button, with a callout box containing the text "Click Next." and the number "4".

Below the "Next" button, there is a "Status" section with "Status: Ready". The "Trunk Type\*" dropdown menu is highlighted with a red box, and a callout box points to it with the text "Select Non-Gatekeeper Controlled ICT." and the number "3". The selected value is "Inter-Cluster Trunk (Non-Gatekeeper Controlled)". The "Device Protocol\*" dropdown menu is also highlighted with a red box and has "Inter-Cluster Trunk" selected.

At the bottom of the form, there is a note: "i \*- indicates required item."

The browser status bar at the bottom shows "Done" and the IP address "192.168.1.10:8443".

# Configuring Non-Gatekeeper Controlled Intercluster Trunks: Configuring Options

The screenshot shows the Cisco CallManager Administration web interface. The page title is "Trunk Configuration - Mozilla Firefox". The browser address bar shows the URL: `https://192.168.1.10:8443/ccmadmin/trunkEdit.do?prod=77&proto=8`. The navigation menu includes "Cisco CallManager Administration" and "Go". The user is logged in as "CCMAdministrator".

The main content area is titled "Trunk Configuration" and includes a "Related Links" section with a "Back To Find/List" button and a "Go" button. The configuration form is divided into two sections: "Status" and "Device Information".

The "Status" section shows "Status: Ready".

The "Device Information" section contains the following fields:

- Product: Inter-Cluster Trunk (Non-Gatekeeper Controlled)
- Device Protocol: Inter-Cluster Trunk
- Device Name\*: NonGK-ICT (highlighted with a red box)
- Description: (empty)
- Device Pool\*: SanJose (highlighted with a red box)
- Call Classification\*: Use System Default
- Media Resource Group List: < None >
- Location\*: Hub\_None
- AAR Group: < None >
- Tunneled Protocol\*: None
- Packet Capture Mode\*: None
- Packet Capture Duration: 0
- Media Termination Point Required

A yellow callout box with a circled "1" points to the "Device Name" and "Device Pool" fields, containing the text: "Specify Name and Device Pool."

The status bar at the bottom shows "Done" and the IP address "192.168.1.10:8443".

# Configuring Non-Gatekeeper Controlled Intercluster Trunks: Configuring Options

Trunk Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccmadmin/trunkEdit.do?key=b7a19929-6c24-17d1-6eb9-0d26114564c1

Enable Inbound FastStart

**Outbound Calls**

Calling Party Selection\* Originator

Calling Line ID Presentation\* Default

Called Party IE Number Type Unknown\* Cisco CallManager

Calling Party IE Number Type Unknown\* Cisco CallManager

Called Numbering Plan\* Cisco CallManager

Calling Numbering Plan\* Cisco CallManager

Caller ID DN

Display IE Delivery

Redirecting Number IE Delivery - Outbound

Enable Outbound FastStart

Codec For Outbound FastStart G711 u-law 64K

**Remote Cisco CallManager Information**

Server 1 IP Address/Host Name\* 192.168.4.1

Server 2 IP Address/Host Name

Server 3 IP Address/Host Name

Save Delete Reset Add New

**2** Specify IP address of remote cluster.

**3** Click Save.

\*- indicates required  
\*\*- Device reset is required

Done 192.168.1.10:8443



# Configuring Gatekeeper Controlled Intercluster Trunks

## Gatekeeper Controlled Intercluster trunk configuration steps:

1. Add Gatekeeper.
2. Add Gatekeeper Controlled ICT.
3. Specify trunk parameters.

# Configuring Gatekeeper Controlled Intercluster Trunks: Add Gatekeeper

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/gateKeeperFindList.do`. The page title is "Find and List Gatekeepers". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", and "User Management". The "Device" dropdown menu is open, showing options: "CTI Route Point", "Gatekeeper", "Gateway", "Phone", "Trunk", and "Device Settings". A yellow callout box with a circled "1" points to the "Gatekeeper" option, containing the text "Select Device > Gatekeeper.". Below the search options, the "Add New" button is highlighted with a red box. A second yellow callout box with a circled "2" points to this button, containing the text "Click Add New.". The search results area shows "No active query. Please enter your search criteria in the search conditions above." The status bar at the bottom of the browser window shows the URL and the IP address "192.168.1.10:8443".

# Configuring Gatekeeper Controlled Intercluster Trunks: Add Gatekeeper

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccadmin/gateKeeperEdit.do`. The page title is "Gatekeeper Configuration" and the user is logged in as "CCMAdministrator".

The "Gatekeeper Configuration" section contains the following information:

- Status:** Status: Ready
- Gatekeeper Information:**
  - Host Name/IP Address\*: 192.168.3.254
  - Description: Gatekeeper
  - Registration Request Time to Live\*: 60
  - Registration Retry Timeout\*: 300
  - Enable Device

Two callout boxes provide instructions:

- 3 Specify Gatekeeper settings.** This callout points to the form fields for Host Name/IP Address, Description, Registration Request Time to Live, and Registration Retry Timeout.
- 4 Click Save** This callout points to the Save button.

# Configuring Gatekeeper Controlled Intercluster Trunks: Add Trunk

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/trunkFindList.do`. The page title is "Find and List Trunks". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Device" menu is expanded, showing options: "CTI Route Point", "Gatekeeper", "Gateway", "Phone", "Trunk", and "Device Settings". The "Trunk" option is highlighted with a red box, and a callout box labeled "1" points to it with the text "Select Device > Trunk.". Below the search options, the "Add New" button is highlighted with a red box, and a callout box labeled "2" points to it with the text "Click Add New.". The search results area contains the text: "No active query. Please enter your search criteria using the options above."

# Configuring Gatekeeper Controlled Intercluster Trunks: Add Trunk

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccadmin/trunkEdit.do?prod=76`. The page title is "Trunk Configuration" and the user is logged in as "CCMAdministrator".

Annotations on the page include:

- A yellow callout box with a circled "4" and the text "Click Next." pointing to a green right-pointing arrow button.
- A yellow callout box with a circled "3" and the text "Select Gatekeeper Controlled ICT." pointing to the "Inter-Cluster Trunk (Gatekeeper Controlled)" option in the "Trunk Type\*" dropdown menu.

The "Trunk Configuration" form contains the following fields:

- Status: Ready
- Trunk Type\*: Inter-Cluster Trunk (Gatekeeper Controlled)
- Device Protocol\*: Inter-Cluster Trunk
- Next button

A note at the bottom left states: "i \*- indicates required item."

# Configuring Gatekeeper Controlled Intercluster Trunks: Configure Trunk

The screenshot displays the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL: `https://192.168.1.10:8443/ccadmin/trunkEdit.do?prod=76&proto=8`. The page title is "Trunk Configuration - Mozilla Firefox".

The interface includes a navigation menu with options: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. A "Log Off" button is also present. The main content area is titled "Trunk Configuration" and includes a "Related Links" section with a "Back To Find/List" button and a "Go" button.

The configuration form is divided into two sections: "Status" and "Device Information".

- Status:** Shows "Status: Ready".
- Device Information:** Contains the following fields:
  - Product: Inter-Cluster Trunk (Gatekeeper Controlled)
  - Device Protocol: Inter-Cluster Trunk
  - Device Name\*: **GK-ICT** (highlighted with a red box and labeled '1')
  - Description: (empty)
  - Device Pool\*: SanJose (dropdown menu)
  - Call Classification\*: Use System Default (dropdown menu)
  - Media Resource Group List: < None > (dropdown menu)
  - Location\*: Hub\_None (dropdown menu)
  - AAR Group: < None > (dropdown menu)
  - Tunneled Protocol\*: None (dropdown menu)
  - Packet Capture Mode\*: None (dropdown menu)
  - Packet Capture Duration: 0
  - Media Termination Point Required

A yellow callout box with a circled "1" points to the "Device Name\*" field, containing the text: "Specify Name and Device Pool."

# Configuring Gatekeeper Controlled Intercluster Trunks

The screenshot shows the 'Trunk Configuration' page in Mozilla Firefox. The browser address bar displays the URL: `https://192.168.1.10:8443/ccmadmin/trunkEdit.do?prod=76&proto=8`. The page contains several configuration sections:

- Inbound Calls:** Includes checkboxes for 'Redirecting Number IE Delivery - Inbound' (checked) and 'Enable Inbound FastStart' (unchecked).
- Outbound Calls:** Includes dropdown menus for 'Calling Party Selection\*' (Originator), 'Calling Line ID Presentation\*' (Default), 'Called Party IE Number Type Unknown\*' (Cisco CallManager), 'Calling Party IE Number Type Unknown\*' (Cisco CallManager), 'Called Numbering Plan\*' (Cisco CallManager), and 'Calling Numbering Plan\*' (Cisco CallManager). It also has a text field for 'Caller ID DN' and checkboxes for 'Display IE Delivery' (checked), 'Redirecting Number IE Delivery - Outbound' (checked), and 'Enable Outbound FastStart' (unchecked). A dropdown for 'Codec For Outbound FastStart' is set to 'G711 u-law 64K'.
- Gatekeeper Information:** Includes dropdown menus for 'Gatekeeper Name\*' (192.168.3.254), 'Terminal Type\*' (Gateway), 'Technology Prefix' (1#), and 'Zone' (sanjose).

Annotations on the screenshot:

- A red box highlights the 'Gatekeeper Information' section.
- A yellow callout box with a circled '2' points to the red box, containing the text: 'Specify previously configured gatekeeper and registration information.'
- A yellow callout box with a circled '3' points to the 'Save' button, containing the text: 'Click Save.'

# Configuring QSIG for Trunks

## **Any Inter-Cluster Trunk can be configured for QSIG:**

- **Done via trunk configuration settings.**
- **Can change existing trunks to QSIG.**



# Configuring QSIG for Trunks

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL: `https://192.168.1.10:8443/ccmadmin/trunkEdit.do?key=b7a19929-6c24-17d1-6eb9-0d26114564c1`. The page title is "Trunk Configuration".

The main configuration area is titled "Inter-Cluster Trunk Configuration". The "Tunneled Protocol\*" dropdown menu is highlighted with a red box and labeled with a circled "1". A callout bubble points to this menu with the text: "Select QSIG as tunneled protocol."

A warning dialog box is open over the configuration area, with the text: "WARNING: Cisco Systems does NOT support the use of protocol profile 0x91 when using trunks with the Tunneled Protocol field set to QSIG. If you set this field to QSIG, the advanced service parameter Protocol Profile 0x91 (ROSE) Encoding must be set to FALSE." The dialog has an "OK" button. A callout bubble labeled with a circled "2" points to the "OK" button with the text: "Verify that service parameters comply with QSIG".

Other configuration fields visible include: "Device Name\*" (NonGK-ICT), "Device Pool\*" (SanJose), "Call Classification\*" (Use System Default), "Location\*" (Hub\_None), and "AAR Group" (<None>).

# Configuring SIP Trunks

## **SIP trunk configuration steps:**

- 1. Add SIP trunk.**
- 2. Specify trunk parameters.**

# Configuring SIP Trunks: Add Trunk

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/trunkFindList.do`. The page title is "Find and List Trunks". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Device" dropdown menu is open, showing options: "CTI Route Point", "Gatekeeper", "Gateway", "Phone", "Trunk", and "Device Settings". The "Trunk" option is highlighted with a red box. A yellow callout box with a circled "1" points to "Trunk" and contains the text "Select Device > Trunk.". Below the search options, the "Add New" button is highlighted with a red box. A yellow callout box with a circled "2" points to "Add New" and contains the text "Click Add New.". The search results area shows "No active query. Please enter your search criteria using the options above."

# Configuring SIP Trunks: Add Trunk

The screenshot displays the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10:8443/ccadmin/trunkEdit.do?prod=95`. The page title is "Trunk Configuration - Mozilla Firefox". The navigation menu includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The main content area shows the "Trunk Configuration" page with a "Status" section indicating "Status: Ready". Below this, there are two dropdown menus: "Trunk Type\*" and "Device Protocol\*", both set to "SIP". A "Next" button is visible below the dropdowns. A yellow callout box with the number "4" and the text "Click Next." points to a green right-pointing arrow icon. Another yellow callout box with the number "3" and the text "Select Non-Gatekeeper Controlled ICT." points to the "Trunk Type\*" dropdown menu. A legend at the bottom left states "\* - indicates required item." The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10:8443".

4 Click Next.

3 Select Non-Gatekeeper Controlled ICT.

\* - indicates required item.

# Configuring SIP Trunks: Configure Trunk

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10:8443/ccmadmin/trunkSave.do`. The page title is "Trunk Configuration - Mozilla Firefox". The navigation menu includes "Cisco CallManager Administration" and "Go". The user is logged in as "CCMAdministrator".

The main content area is titled "Trunk Configuration" and includes a "Related Links" section with a "Back To Find/List" link. Below this is a "Status" section with an "Add successful" message. The "Device Information" section contains the following fields:

- Product: SIP Trunk
- Device Protocol: SIP
- Device Name\*: SIP-Trunk
- Description: [Empty text box]
- Device Pool\*: SanJose
- Call Classification\*: Use System Default
- Media Resource Group List: < None >
- Location\*: Hub\_None
- AAR Group: < None >
- Packet Capture Mode\*: None
- Packet Capture Duration: 0

At the bottom of the form, there are two checkboxes: "Media Termination Point Required" (unchecked) and "Retry Video Call as Audio" (checked). A red box highlights the "Device Name\*" and "Device Pool\*" fields, and a callout box with the number "1" points to them, containing the text "Specify Name and Device Pool."

# Configuring SIP Trunks: Configure Trunk

**Outbound Calls**

Calling Party Selection\*

Calling Line ID Presentation\*

Calling Name Presentation\*

Caller ID DN

Caller Name

Redirecting Diversion Header Delivery - Outbound

**SIP Information**

Destination Address\*

Destination Address is an SRV

Destination Port\*

MTP Preferred Originating Codec\*

Presence Group\*

SIP Trunk Security Profile\*

Rerouting Calling Search Space

Out-Of-Dialog Refer Calling Search Space

SUBSCRIBE Calling Search Space

SIP Profile\*

DTMF Signaling Method

Done

192.168.1.10:8443

**2** Specify SIP information.

**3** Click Save.

# Summary

- **Trunks are used for intercluster signalling and interworking with other VoIP networks.**
- **Non-Gatekeeper Controlled Intercluster Trunks are configured between two CallManager clusters.**
- **Gatekeeper Controlled Intercluster Trunks are configured to use gatekeepers to interconnect multiple clusters.**
- **QSIG Intercluster Trunks are configured to provide supplementary services between clusters.**
- **SIP Trunks are configured to interconnect with SIP networks.**

# CISCO SYSTEMS







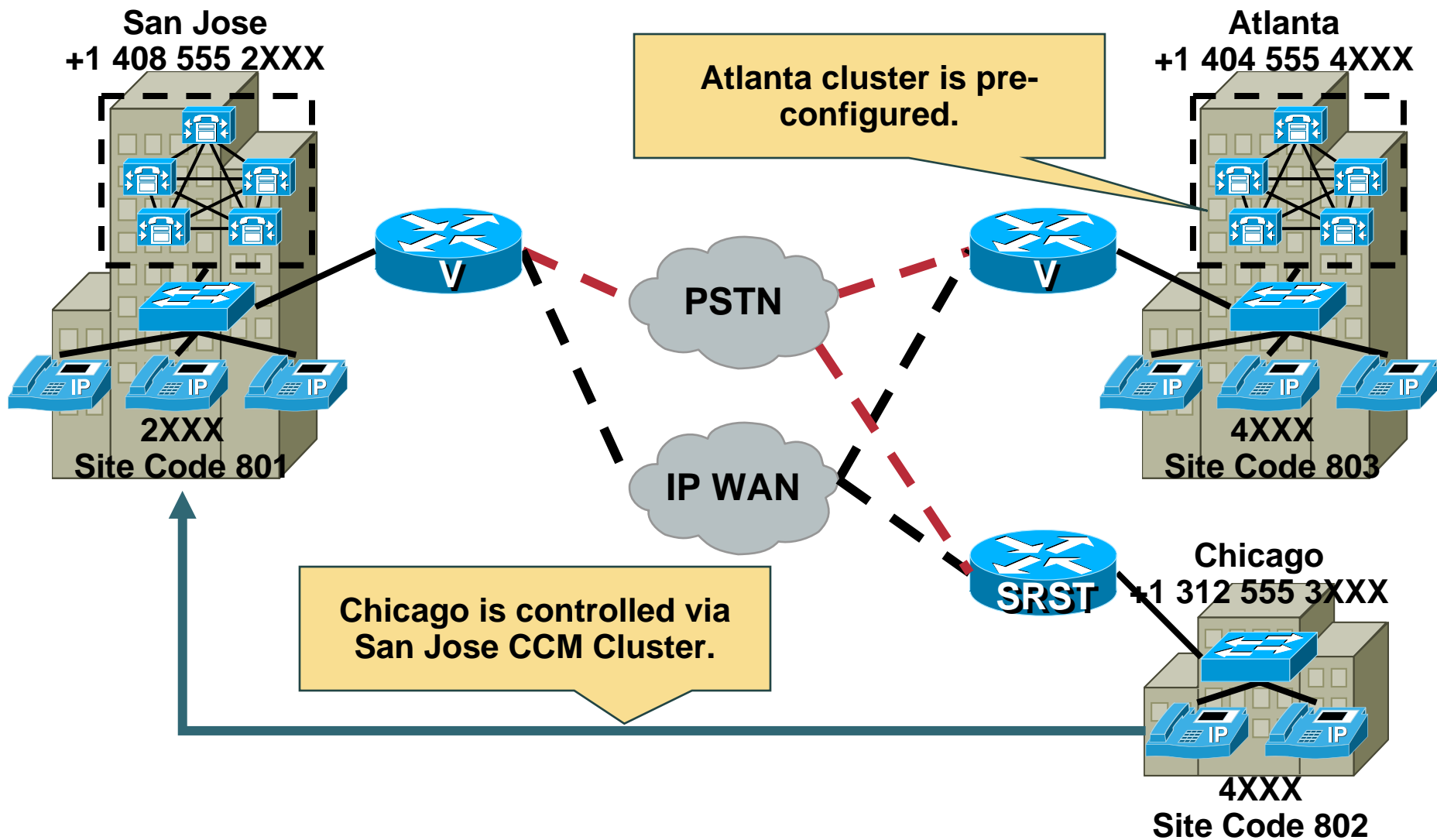
## Deploying a Dial Plan on CallManager 5.0

# Configuring a Dial Plan

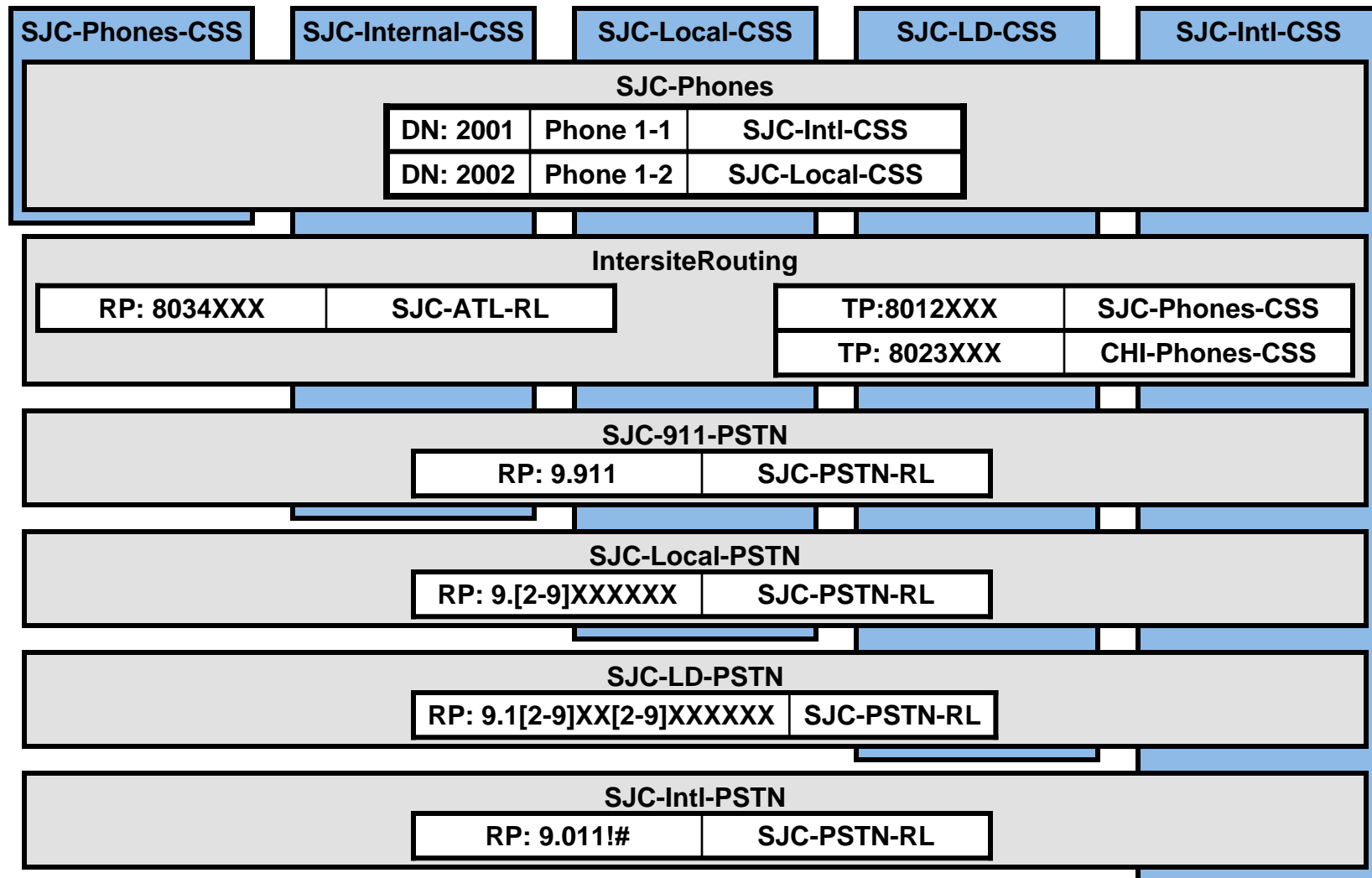
# Objectives

- **Dial Plan Scenario**
- **Configuring Intrasite Routing**
- **Configuring On-Cluster Intersite Routing**
- **Configuring Off-Cluster Intersite Routing**
- **Configuring PSTN Routing and Calling Privileges**

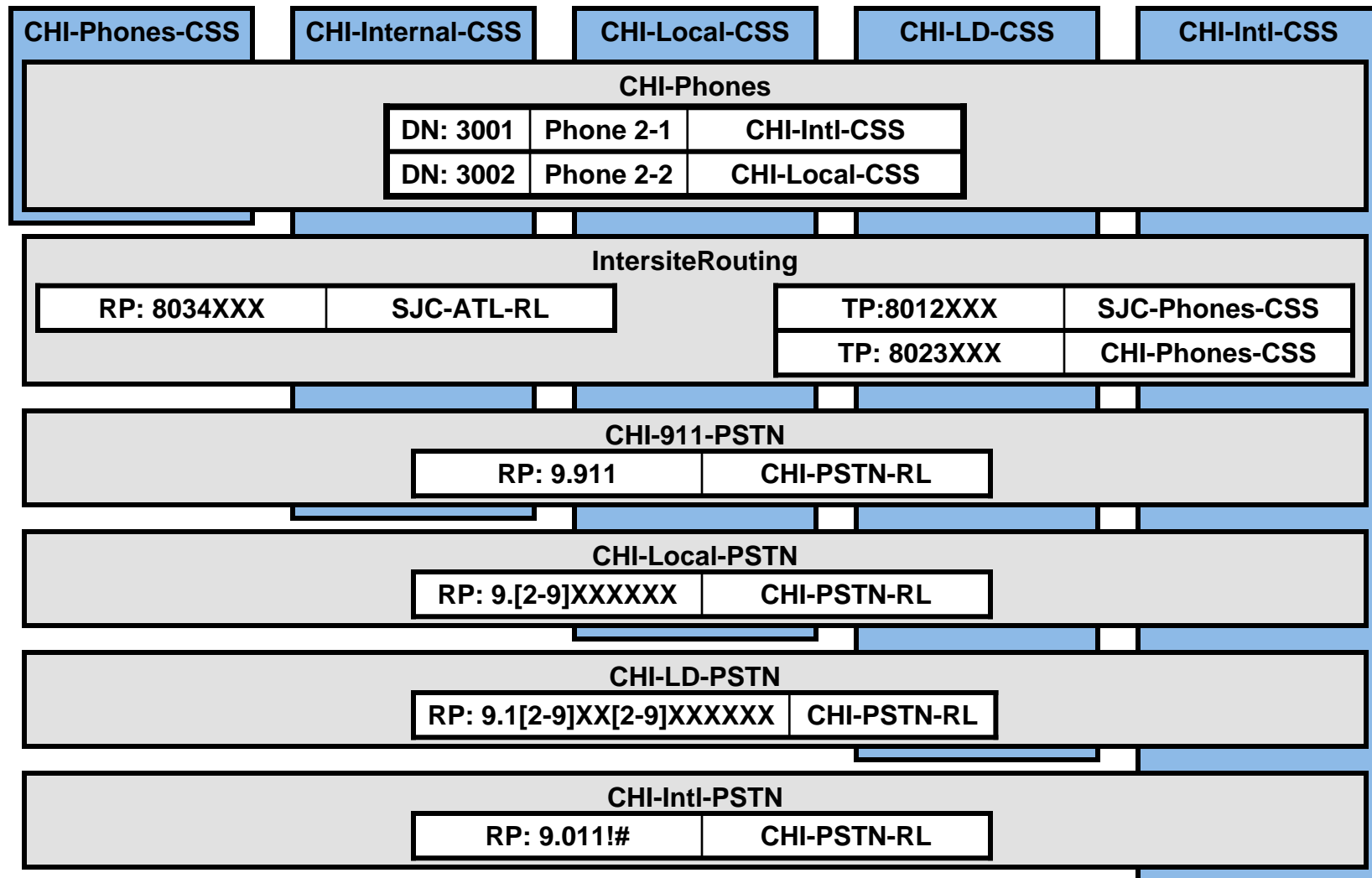
# Dial Plan Scenario



# Dial Plan Scenario: San Jose Partitions and CSSs



# Dial Plan Scenario: Chicago Partitions and CSSs



# Dial Plan Scenario: RPs and TPs

Pattern	Partition	Type	Destination	Digit Manipulation	Description
<b>Intersite Routing</b>					
8012XXX	IntersiteRouting	TP	SJC-Phones-CSS	Called: 2XXX	Intersite Routing
8023XXX	IntersiteRouting	TP	CHI-Phones-CSS	Called: 3XXX	Intersite Routing
8034XXX	IntersiteRouting	RP	ATL-RL	NA	Intersite Routing
<b>PSTN Routing</b>					
911	SJC-911-PSTN	RP	SJC-PSTN-RL	Called: PreDot	Emergency Calls
9.911	CHI-911-PSTN	RP	CHI-PSTN-RL	Calling: ExMask	
9.[2-9]XXXXXX	SJC-Local-PSTN	RP	SJC-PSTN-RL	Called: PreDot	Local PSTN Routing
	CHI-Local-PSTN	RP	CHI-PSTN-RL	Calling: ExMask	
9.1[2-9]XX[2-9]XXXXXX	SJC-LD-PSTN	RP	SJC-PSTN-RL	Called: PreDot	LD PSTN Routing
	CHI-LD-PSTN	RP	CHI-PSTN-RL	Calling: ExMask	
9.011!	SJC-Intl-PSTN	RP	SJC-PSTN-RL	Called: PreDot	Intl PSTN Routing
9.011!#	CHI-Intl-PSTN	RP	CHI-PSTN-RL	Calling: ExMask	

# Dial Plan Scenario: RGs and RLs

Route List	Route Groups	Gateway/Trunk	Digit Manipulation	Description
ATL-RL	ATL-Trunk SJC-PSTN	NonGK-ICT Router1	NA Called: 14045552XXX	Route List for Atlanta Calls.
SJC-PSTN-RL	SJC-PSTN	Router1	NA	Route List for SJC PSTN Calls
CHI-PSTN-RL	CHI-PSTN	192.168.2.254	NA	Route List for CHI PSTN Calls

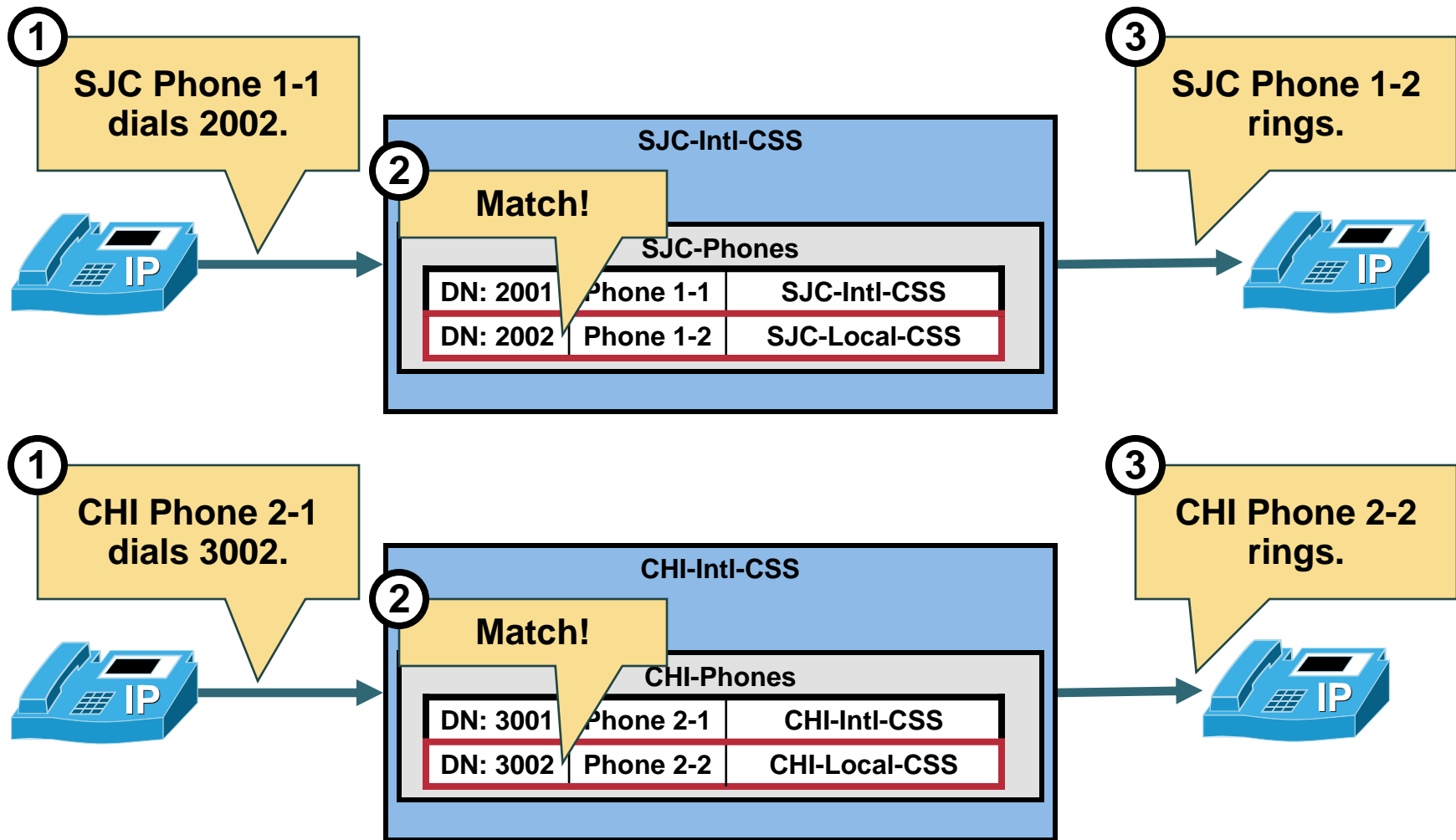
# Configuring Intrasite Routing

**Intrasite routing is done via CSSs, Partitions and Translation Patterns:**

- 1. Review required dial plan components.**
- 2. Configure Partitions.**
- 3. Configure Calling Search Spaces**
- 4. Assign phone DNs to correct partitions and configure CSSs.**

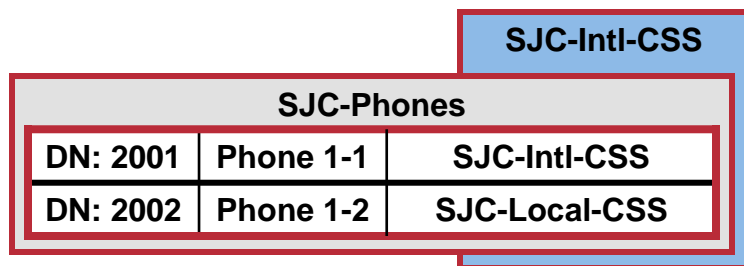


# Configuring Intrasite Routing

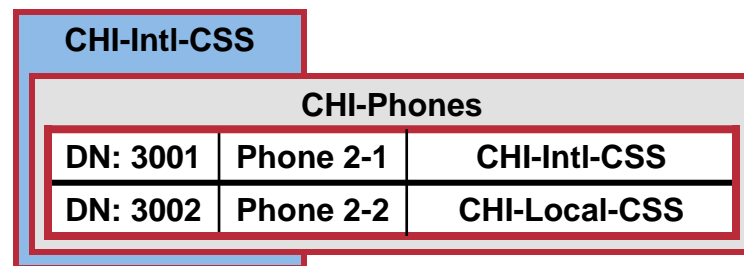


# Configuring Intrasite Routing

## San Jose



## Chicago



# Configuring Intrasite Routing

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/partitionFindList.do`. The page title is "Find and List Partitions". The navigation menu is open, showing the path: **Call Routing > Class of Control > Partition**. A yellow callout box with a pointer to the "Partition" option contains the text: **Select Call Routing > Class of Control > Partition and Click Add New**. The interface also shows a search bar with "begins with" and a "Find" button, and a "Log Off" button in the top right corner.

# Configuring Intrasite Routing

The screenshot shows the Cisco CallManager Administration interface in a Mozilla Firefox browser window. The page title is "Partition Configuration" and the URL is "https://192.168.1.10:8443/ccmadmin/partitionEdit.do". The user is logged in as "CCMAdministrator".

Annotations on the page:

- 1**: A yellow callout box with the text "Save configuration." points to a save icon in the top left of the configuration area.
- 2**: A yellow callout box with the text "Configure multiple partitions, comma delimits description." points to a text input field containing the following text:  
SJC-Phones, San Jose Phones  
CHI-Phones, Chicago Phones

The page also includes a "Status" section showing "Status: Ready" and a "Related Links" section with a "Back To Find/List" button. A "Done" status is shown at the bottom left, and the IP address "192.168.1.10:8443" is shown at the bottom right.

# Configuring Intrasite Routing

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/cssFindList.do`. The page title is "Find and List Calling Search Spaces - Mozilla Firefox". The navigation menu is open, showing the path: **Call Routing > Class of Control > Calling Search Space**. A yellow callout box highlights this path with the text: "Select Call Routing > Class of Control > Calling Search Space". The main content area shows a search form with a "Find" button and a "Search Within Results" checkbox. The status bar at the bottom of the browser window shows the IP address `192.168.1.10:8443`.

# Configuring Intrasite Routing

**3** Save configuration.

**1** Specify CSS name.

**2** Add appropriate partition from list.

Calling Search Space Configuration - Mozilla Firefox  
https://192.168.1.10:8443/ccadmin/cssEdit.do?clone=1&key=750ef8df-142a-9061-cbb2-db0c7b6e0979

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as:CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Calling Search Space Status: Ready

Calling Search Space Information

Name\* SJC-Intl-CSS

Description

Route Partitions for this Calling Search Space

Available Partitions

Selected Partitions (Ordered by highest priority)

SJC-Phones

Done 192.168.1.10:8443

# Configuring Intrasite Routing

Directory Number Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccadmin/directoryNumberEdit.do?key=66d90053-935f-ec72-c061-479153551f8a&mapkey=74e26334-46a1-73c7-0d57-c8

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Directory Number Configuration Related Links: Back To Find/List Go

Status

Status: Ready

Note: Changes to Line or Directory Number settings require restart.

Directory Number Information

Directory Number*	2001
Route Partition	SJC-Phones
Description	SJC 2001
Alerting Name	Max Miller
ASCII Alerting Name	Max Miller

Allow Control of Device from CTI

Associated Devices

SEP00123F19CBD6	Edit Device
	Edit Line Appearance

Dissociate Devices

Done

192.168.1.10:8443

**1 Add phone DN to appropriate partition.**

# Configuring Intrasite Routing

Directory Number Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccadmin/directoryNumberEdit.do?key=66d90053-935f-ec72-c061-479153551f8a&mapkey=74e26334-46a1-73c7-0d57-c8

▼ ▲

Dissociate Devices

**Directory Number Settings**

Voice Mail Profile < None >

Calling Search Space **SJC-Intl-CSS**

Presence Group\* Standard Presence group

AAR Group < None >

User Hold Audio Source < None >

Network Hold Audio Source < None >

Auto Answer\* Auto Answer Off

**Call Forward and Call Pickup Settings**

	Voice Mail Destination	Calling Search Space
Forward All	<input type="checkbox"/> or	< None >
Secondary Calling Search Space for Forward All		< None > Find
Forward Busy Internal	<input type="checkbox"/> or	< None >
Forward Busy External	<input type="checkbox"/> or	< None >
Forward No Answer Internal	<input type="checkbox"/> or	< None >
Forward No Answer External	<input type="checkbox"/> or	< None >
Forward No Coverage Internal	<input type="checkbox"/> or	< None >

Done 192.168.1.10:8443

**2** Select appropriate CSS and save.

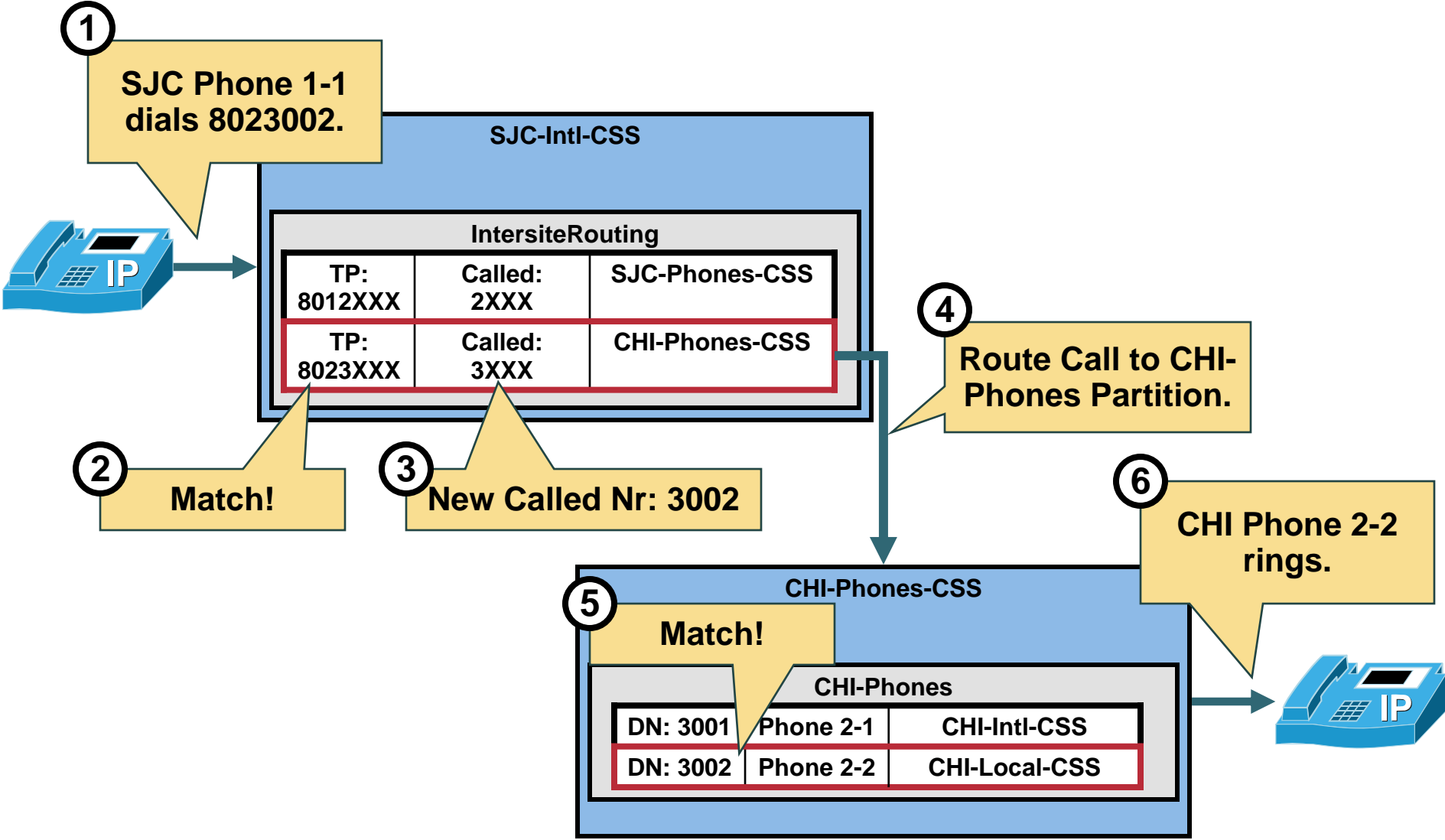


# Configuring On-Cluster Intersite Routing

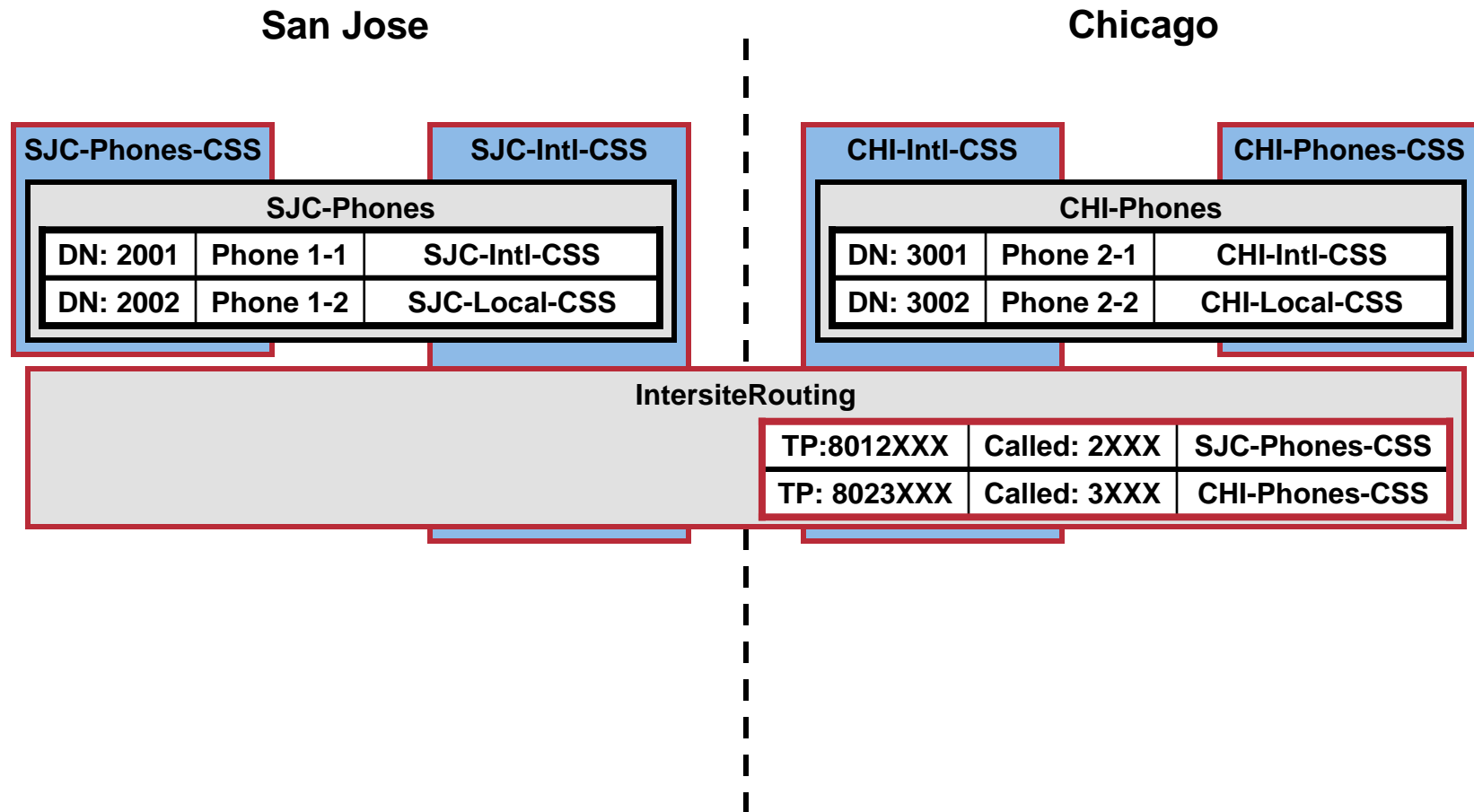
## **Intersite Routing configuration steps:**

- 1. Review required dial-plan components.**
- 2. Add additional partitions and CSSs.**
- 3. Configure Translation Patterns for on-cluster routing.**

# Configuring On-Cluster Intersite Routing



# Configuring On-Cluster Intersite Routing



# Configuring On-Cluster Intersite Routing

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/translationFindList.do`. The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". A navigation menu is visible with the following items: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "Call Routing" menu is expanded, showing a list of options: AAR Group, Dial Rules, Route Filter, Route/Hunt, SIP Route Pattern, Class of Control, Client Matter Codes, Forced Authorization Codes, Translation Pattern (highlighted), Call Park, Call Pickup Group, Directory Number, Meet-Me Number/Pattern, Route Plan Report, and Dial Plan Installer. A yellow callout box with a black border points to the "Translation Pattern" option, containing the text: "Go to Call Routing > Translation Pattern and add a new pattern." Below the navigation menu, there is a search area with a "Find" button and a "Search Within Results" checkbox. The status bar at the bottom of the browser window shows the URL `https://192.168.1.10:8443/ccmadmin/translationFindList.do` and the IP address `192.168.1.10:8443`.

# Configuring On-Cluster Intersite Routing

The screenshot displays the Cisco CallManager Administration web interface for configuring a Translation Pattern. The browser window title is "Translation Pattern Configuration - Mozilla Firefox" and the URL is "https://192.168.1.10:8443/ccadmin/translationEdit.do?clone=1&key=b98e452f-ec8e-ad87-7266-a6926a0c575b". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu shows "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help".

The main content area is titled "Translation Pattern Configuration" and contains the following fields and callouts:

- 1 Specify Translation Pattern.** Points to the "Translation Pattern" field, which contains "8012XXX".
- 2 Specify Partition** Points to the "Partition" field, which contains "Intersite-Routing".
- 3 Specify CSS which will be used by the TP.** Points to the "Calling Search Space" dropdown menu, which is set to "SJC-Phones-CSS".

Other fields in the "Pattern Definition" section include "Description", "Numbering Plan" (set to "< None >"), "Route Filter" (set to "< None >"), "MLPP Precedence\*" (set to "Default"), and "Route Option" (with "Route this pattern" selected and "Block this pattern" set to "No Error"). Checkboxes for "Provide Outside Dial Tone" and "Urgent Priority" are both checked. The "Calling Party Transformations" section includes a checkbox for "Use Calling Party's External Phone Number Mask" and a "Calling Party Transform Mask" field.

# Configuring On-Cluster Intersite Routing

The screenshot shows the 'Translation Pattern Configuration' page in Mozilla Firefox. The browser address bar shows the URL: `https://192.168.1.10:8443/ccadmin/translationEdit.do?clone=1&key=b98e452f-ec8e-ad87-7266-a6926a0c575b`. The page contains several configuration sections:

- MLPP Precedence\***: Default
- Calling Search Space**: SJC-Phones-CSS
- Route Option**:  Route this pattern,  Block this pattern (No Error)
- Provide Outside Dial Tone,  Urgent Priority
- Calling Party Transformations**:
  - Use Calling Party's External Phone Number Mask
  - Calling Party Transform Mask: [Empty]
  - Prefix Digits (Outgoing Calls): [Empty]
  - Calling Line ID Presentation\*: Default
  - Calling Name Presentation\*: Default
- Connected Party Transformations**:
  - Connected Line ID Presentation\*: Default
  - Connected Name Presentation\*: Default
- Called Party Transformations**:
  - Discard Digits: <None >
  - Called Party Transform Mask: 2XXX
  - Prefix Digits (Outgoing Calls): [Empty]

Annotations:

- 4**: Configure Digit Manipulation. (Points to the top of the configuration page)
- No Calling Party Transformation required.** (Points to the Calling Party Transformations section)
- No Connected Party Transformation required.** (Points to the Connected Party Transformations section)
- Called Party Transform Mask 2XXX.** (Points to the Called Party Transform Mask field)
- 5**: Save configuration. (Points to the Save button)

# Configuring On-Cluster Intersite Routing

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/rprptFindList.do`. The page title is "Route Plan Report" and the user is logged in as "CCMAdministrator".

**Navigation:** Cisco CallManager Administration



**System:** Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, Help

**Route Plan Report** Related Links: View in file, Go

**Status:** 2 records found

**Search Options:** Find Translation Pattern where Pattern/Directory Number begins with Find

**Search Results:**

	Pattern/Directory Number	Partition	Type	Route Detail
	<a href="#">8012XXX</a>	<a href="#">Intersite-Routing</a>	Translation Pattern	
	<a href="#">8023XXX</a>	<a href="#">Intersite-Routing</a>	Translation Pattern	

Rows per Page: 25

Done 192.168.1.10:8443

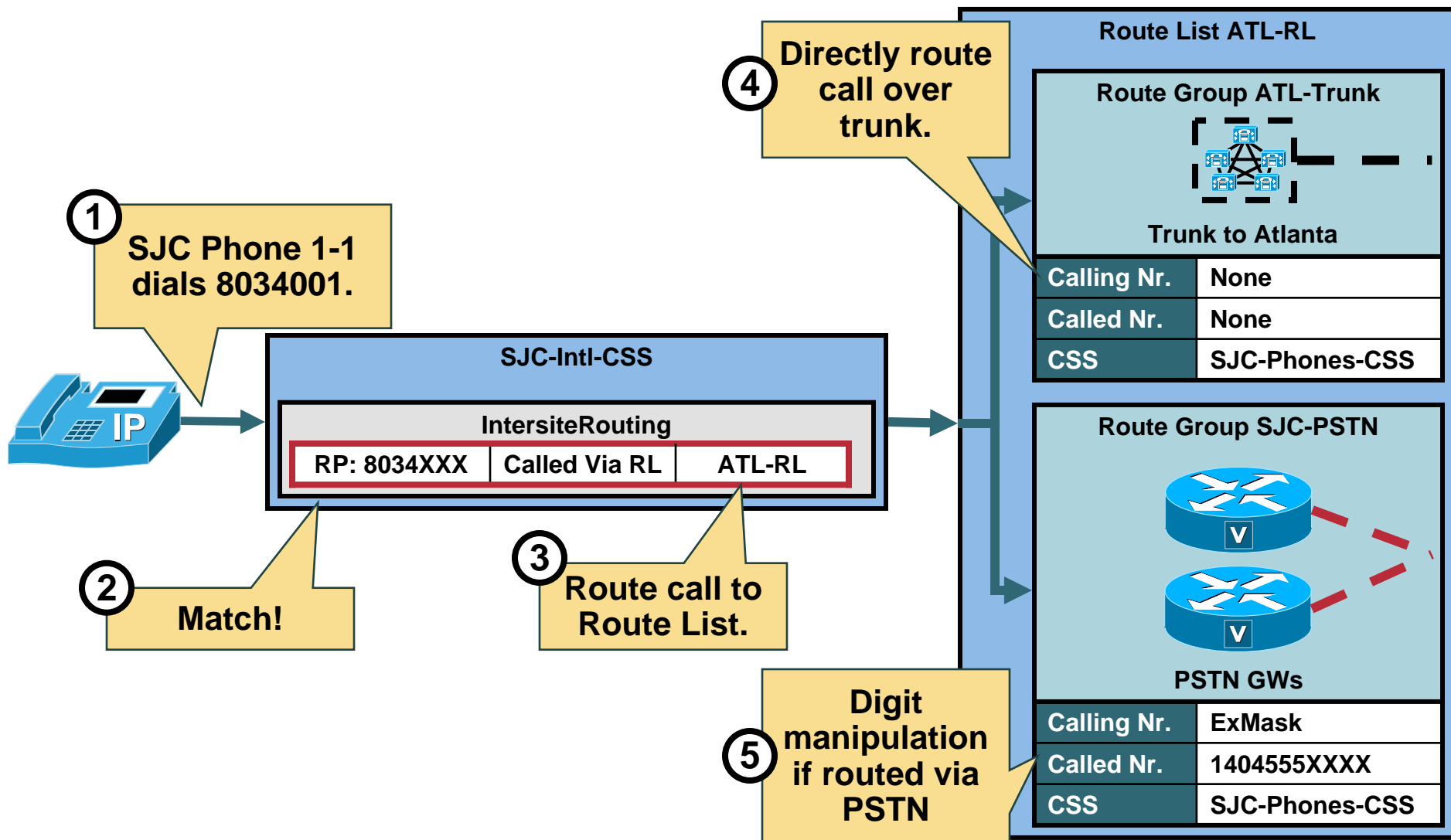
# Configuring Off-Cluster Intersite Routing

## **Intersite Routing configuration steps:**

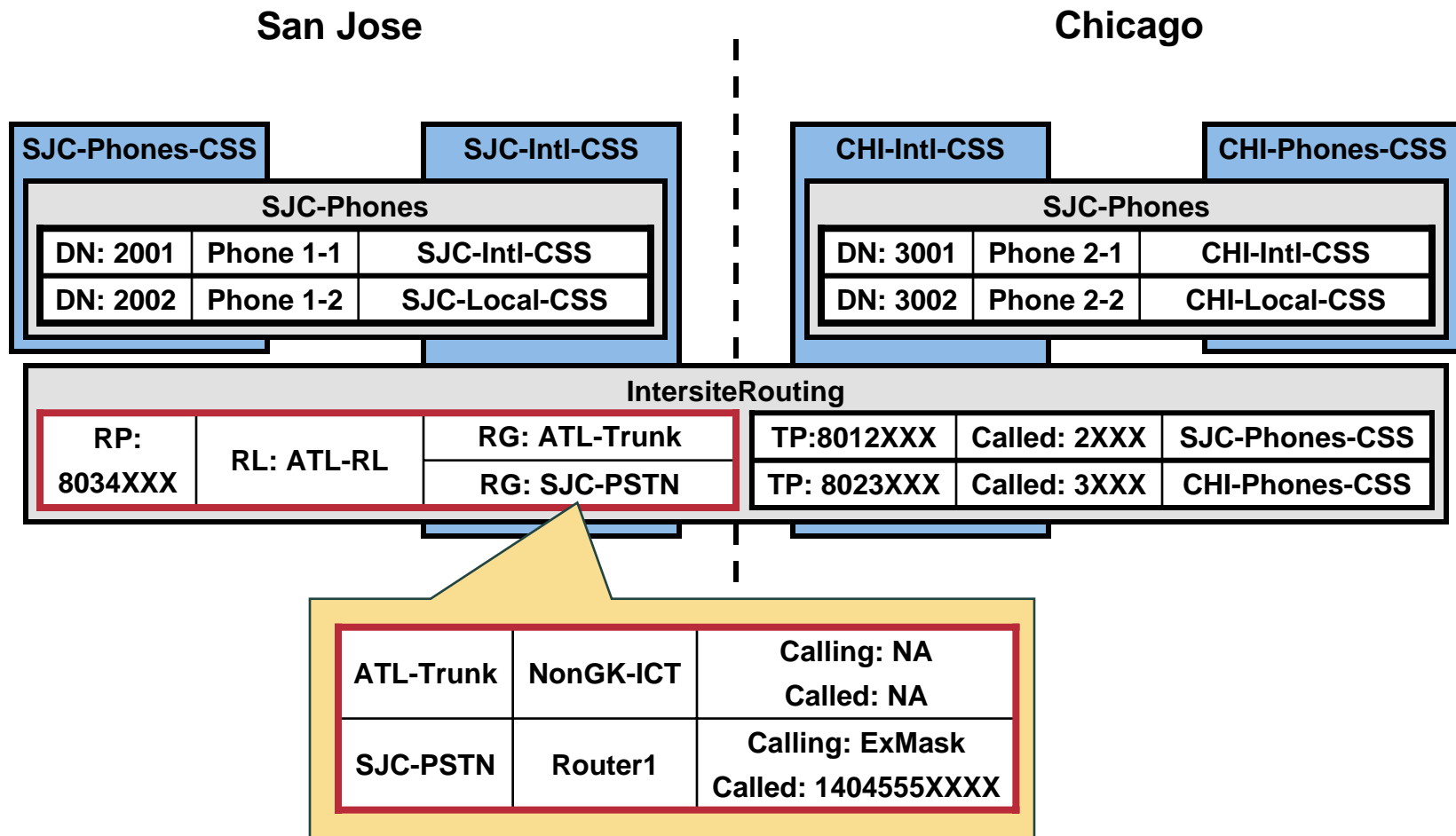
- 1. Review required dial-plan components.**
- 2. Configure Route Groups and Route Lists.**
- 3. Configure Route Patterns.**



# Configuring Off-Cluster Intersite Routing



# Configuring Off-Cluster Intersite Routing



# Configuring Off-Cluster Intersite Routing: Adding Route Groups

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays the URL: <https://192.168.1.10:8443/ccmadmin/routeGroupFindList.do?lookup=false&multiple=true&recCnt=0&colCnt=2>. The page title is "Find and List Route Groups - Mozilla Firefox". The navigation menu is open, showing the path: **Call Routing > Route/Hunt > Route Group**. A yellow callout box with a black border and a pointer to the "Route Group" option contains the text: "Go to Call Routing > Route/Hunt > Route Group and add a new Route Group". The main content area shows a table with columns for "Route Group Name" and "Route Group Description". The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10:8443".

# Configuring Off-Cluster Intersite Routing: Configuring Route Groups

The screenshot shows the 'Route Group Configuration' web interface in Mozilla Firefox. The interface is divided into several sections:

- Route Group Information:** Contains fields for 'Route Group Name\*' (set to 'ATL-Trunk') and 'Distribution Algorithm\*' (set to 'Circular').
- Route Group Member Information:** Includes a search box for 'Device Name contains', a list of 'Available Devices' (192.168.1.254, 192.168.2.254, GK-ICT, NonGK-ICT, S1/DS1-0@Router1), a 'Port(s)' dropdown (set to 'None Available'), and an 'Add to Route Group' button.
- Current Route Group Members:** Shows a list of 'Selected Devices (ordered by highest priority)' with 'NonGK-ICT (All Ports)' selected. It also includes a 'Reverse Order of Selected Devices' button.

Four numbered callouts provide instructions:

- 1:** Specify a name for the Route Group. This Route Group will point to ATL. (Points to the 'Route Group Name\*' field)
- 2:** Select the Trunk/Gateway you want to add to the Route Group. In this case we select the NonGK-ICT to ATL. (Points to the 'NonGK-ICT' device in the 'Available Devices' list)
- 3:** Click Add to Route Group to add selected Trunk/Gateway. (Points to the 'Add to Route Group' button)
- 4:** Save configuration. (Points to the 'Save' button in the top navigation bar)

# Configuring Off-Cluster Intersite Routing: Configuring Route Groups

**Route Group Configuration - Mozilla Firefox**

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccadmin/routeGroupSave.do

**Route Group Configuration** Related Links: Back To Find/List Go

**Status**  
Add successful

**Route Group Information**

Route Group Name\* SJC-PSTN

Distribution Algorithm\* Circular

**Route Group Member Information**

**Find Devices to Add to Route Group**

Device Name contains  Find

Available Devices (select device, then select port below)

- 192.168.1.254
- 192.168.2.254
- GK-ICT
- NonGK-ICT
- S1/DS1-0@Router1**

Port(s) None Available

Add to Route Group

**Current Route Group Members**

Selected Devices\* (ordered by highest priority)

- S1/DS1-0@Router1 (All Ports)**

Reverse Order of Selected Devices

Done 192.168.1.10:8443

**Here we add another RG which contains the SJC PSTN GW.**

# Configuring Off-Cluster Intersite Routing: Adding Route Lists

The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator". A navigation menu is open, showing the path: **Call Routing > Route/Hunt > Route List**. A yellow callout box with a circled "1" points to the "Route List" option and contains the text: "Go to Call Routing > Route/Hunt > Route List and add a new Route List." The browser address bar at the bottom shows `https://192.168.1.10:8443/ccmadmin/routeListTwoFindList.do`.

# Configuring Off-Cluster Intersite Routing: Adding Route Lists

**Route List Configuration - Mozilla Firefox**  
File Edit View Go Bookmarks Tools Help  
https://192.168.1.10:8443/ccmadmin/routeListTwoEdit.do

**Cisco CallManager Administration** For Cisco IP Telecommunication Solutions  
Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

**Route List Configuration** Related Links: Back To Find/List Go

**Status**  
Status: Ready

**Route List Information**  
Name\* ATL-RL  
Description  
Cisco CallManager Group\* Default

Save

\* - indicates required item.

**2** Specify the name of the Route List.

**3** Select the Cisco CallManager Group which should handle the RL.

**4** Save the configuration.

Done 192.168.1.10:8443

# Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10:8443/ccmadmin/routeListTwoSave.do`. The page title is "Route List Configuration". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The "Route List Configuration" section includes a "Status" area with a message "Add successful". Below this is the "Route List Information" section with the following fields:

- Name\*: ATL-RL
- Description: (empty)
- Cisco CallManager Group\*: Default
- Enable this Route List (change effective on Save; no reset required)

The "Route List Member Information" section contains an "Add Route Group" button, which is highlighted by a yellow callout box with a circled "1" and the text "Click Add Route Group.". Below the button is a list of "Selected Groups \* (ordered by highest priority)" and a "Removed Groups (Save)" section.



# Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists

The screenshot displays the Cisco CallManager Administration web interface for configuring a Route List. The browser window title is "Route List Detail Configuration - Mozilla Firefox". The address bar shows the URL: <https://192.168.1.10:8443/ccmadmin/routeListTwoDetailEdit.do?devkey=05a661fa-a272-6413-60e4-b1ee07f45f87&order=;0>. The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu shows "System", "Call Routing", "Application", "User Management", "Bulk Administration", and "Help". The "Route List Detail Conf" page has a "Save configuration." button (labeled 3) and a "Back To Find/List" link. The "Status" section shows "Status: Ready". The "Route List Member Information" section has a "Route Group \*" dropdown menu (labeled 2) with the value "ATL-Trunk-[NON-QSIG]". The "Calling Party Transformations" section has a "Use Calling Party's External Phone Number Mask\*" dropdown menu with the value "Off". The "Called Party Transformations" section has a "Discard Digits" dropdown menu with the value "< None >". A "Save" button is located at the bottom left of the configuration area. The status bar at the bottom shows "Done" and the IP address "192.168.1.10:8443".

# Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10:8443/ccmadmin/routeListTwoSave.do`. The page title is "Route List Configuration". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The "Route List Configuration" section includes a "Status" area with an "Update successful" message. Below this is the "Route List Information" section with the following fields:

- Name\*: SJC-ATL-RL
- Description: (empty)
- Cisco CallManager Group\*: Default
- Enable this Route List (change effective on Save; no reset required)

The "Route List Member Information" section contains an "Add Route Group" button. A yellow callout box with a circled "4" points to this button, containing the text: "Click Add Route Group to add another RG." Below the button, there are two lists of route groups. The "Selected Groups" list (ordered by highest priority) contains "ATL-Trunk". The "Removed Groups" list is currently empty.

# Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists

The screenshot displays the Cisco CallManager Administration web interface for configuring a Route List. The browser window title is "Route List Detail Configuration - Mozilla Firefox". The URL is [https://192.168.1.10:8443/ccmadmin/routeListTwoDetailEdit.do?devkey=aeafac59-8c56-e32a-5342-b5ac8e8ef866&order=ROUTE\\_GROUP;1](https://192.168.1.10:8443/ccmadmin/routeListTwoDetailEdit.do?devkey=aeafac59-8c56-e32a-5342-b5ac8e8ef866&order=ROUTE_GROUP;1). The page is titled "Route List Detail Configuration" and shows the configuration for a route list member. The "Route Group" is set to "SJC-PSTN-[NON-QSIG]". The "Calling Party Transformations" section has "Use Calling Party's External Phone Number Mask" set to "On". The "Called Party Transformations" section has "Called Party Transform Mask" set to "1404555XXXX".

**8** Save configuration.

**5** Select the SJC-PSTN RG.

**6** Select On for External Phone Number Mask.

**7** Configure a Called Party Transform Mask of 1404555XXXX (ATL PSTN Nr.)

# Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists

**Reset RL to ensure it registers with CCM.** 9

**Route List Configuration**

**Status**  
Update successful

**Route List Information**

Name\* ATL-RL  
Description  
Cisco CallManager Group\* Default  
 Enable this Route List (change effective on Save; no reset required)

**Route List Member Information**

**Add Route Group**

Selected Groups \* (ordered by highest priority)

- ATL-Trunk
- SJC-PSTN

Removed Groups (to be removed from Route List when you click Save)

Done

# Configuring Off-Cluster Intersite Routing: Configuring Route Patterns

The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser. The browser's address bar displays `https://192.168.1.10:8443/ccmadmin/showHome.do`. The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". A navigation menu is visible with the following items: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "Call Routing" menu is expanded, showing a list of options: AAR Group, Dial Rules, Route Filter, Route/Hunt, SIP Route Pattern, Class of Control, Client Matter Codes, Forced Authorization Codes, Translation Pattern, Call Park, Call Pickup Group, Directory Number, Meet-Me Number/Pattern, Route Plan Report, and Dial Plan Installer. The "Route/Hunt" option is selected, and its sub-menu is open, listing: Route Group, Route List, Route Pattern, Line Group, Hunt List, and Hunt Pilot. The "Route Pattern" option is highlighted. A yellow callout box with a circled "1" contains the text: "Go to Call Routing > Route/Hunt > Route Pattern and add a new Route Pattern." The browser's status bar at the bottom shows the URL `https://192.168.1.10:8443/ccmadmin/routePattern2FindList.do` and the IP address `192.168.1.10:8443`.

# Configuring Off-Cluster Intersite Routing: Configuring Route Patterns

**Specify Route Pattern.** ①

**Select the appropriate partition.** ③

**Select the Numbering Plan.** ④

**Select previously configured RL.** ⑤

②

Route Pattern Configuration - Mozilla Firefox  
https://192.168.1.10:8443/ccadmin/routePattern2Edit.do?dialplan=1&key=714eaff8-55c2-11d3-b52d-0008c7c6a4c6  
Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator  
System Call Routing Media Resources Voice Management Bulk Administration Help Log Off  
Route Pattern Configuration Related Links: Back To Find/List Go  
Pattern Definition  
Route Pattern\* 803.4XXX  
Route Partition Intersite-Routing  
Description  
Numbering Plan\* NANP  
Route Filter < None >  
MLPP Precedence\* Default  
Gateway/Route List\* ATL-RL (Edit)  
Route Option  
 Route this pattern  
 Block this pattern No Error  
Call Classification\* OffNet  
 Allow Device Override  Provide Outside Dial Tone  Allow Overlap Sending  
 Require Forced Authorization Code  
Authorization Level\* 0  
 Require Client Matter Code  
Calling Party Transformations  
Done 192.168.1.10:8443

# Configuring Off-Cluster Intersite Routing: Configuring Route Patterns

Route Pattern Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccmadmin/routePattern2Edit.do?key=0e5ca0d9-f2bd-4cbc-9c8c-0f8f727b1411

Require Client Matter Code

**Calling Party Transformations**

Use Calling Party's External Phone Number Mask

Calling Party Transform Mask

Prefix Digits (Outgoing Calls)

Calling Line ID Presentation\*

Calling Name Presentation\*

**Connected Party Transformations**

Connected Line ID Presentation\*

Connected Name Presentation\*

**Called Party Transformations**

Discard Digits

Called Party Transform Mask

Prefix Digits (Outgoing Calls)

**ISDN Network-Specific Facilities Information Element**

Network Service Protocol

Carrier Identification Code

Network Service	Service Parameter Name	Service Parameter Value
<input type="text" value="- Not Selected -"/>	<input type="text" value="list &gt;"/>	<input type="text"/>

Done

192.168.1.10:8443

**6** Configure digit manipulation here or use Route Lists for digit manipulation.

**7** Save configuration.

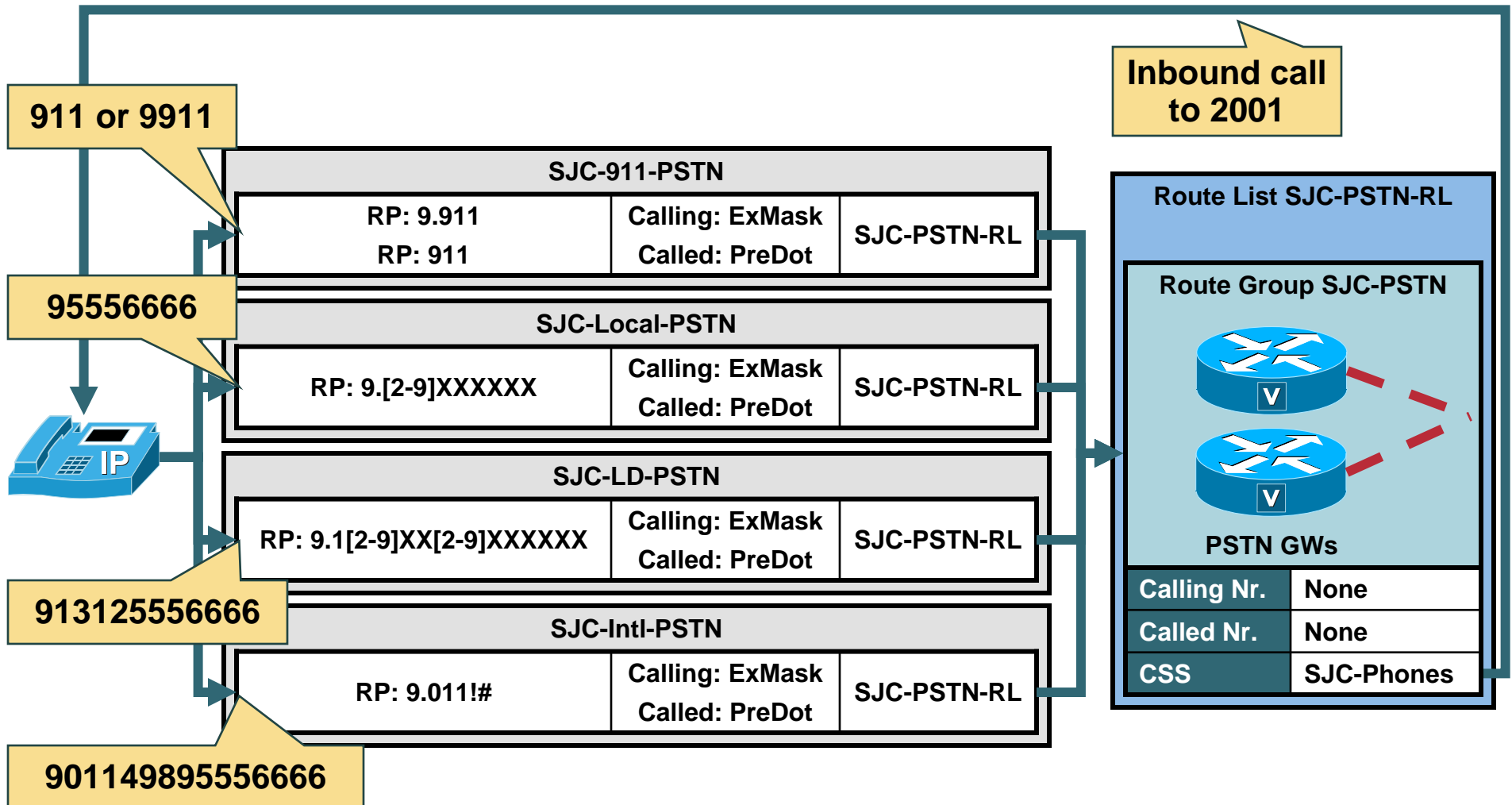
# Configuring PSTN Routing and Calling Privileges

## **PSTN Routing and Calling Privileges configuration steps:**

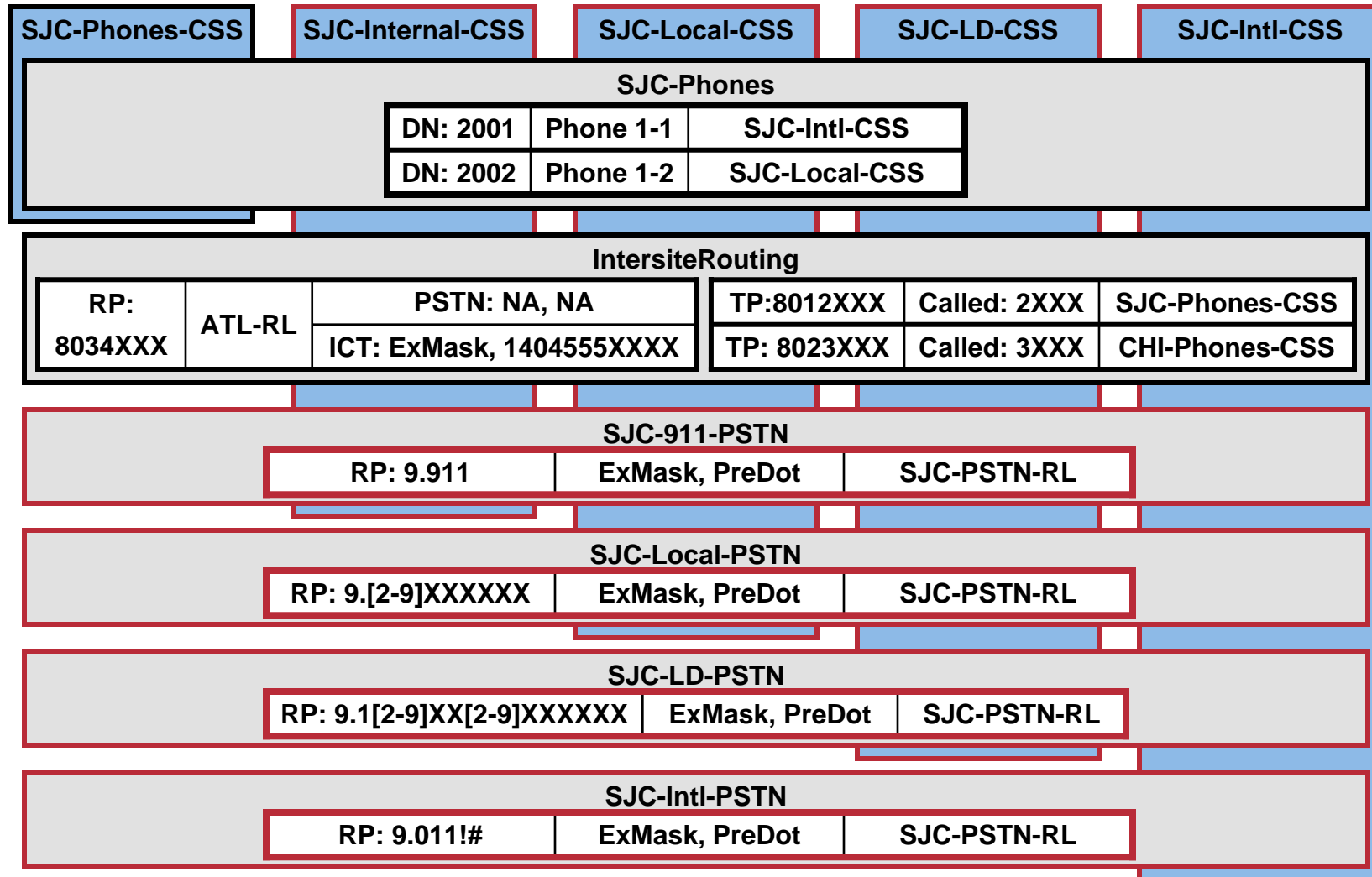
- 1. Review required dial-plan components.**
- 2. Add additional partitions and CSSs.**
- 3. Add additional route lists.**
- 4. Configure additional route patterns.**
- 5. Assign inbound CSSs to gateways.**



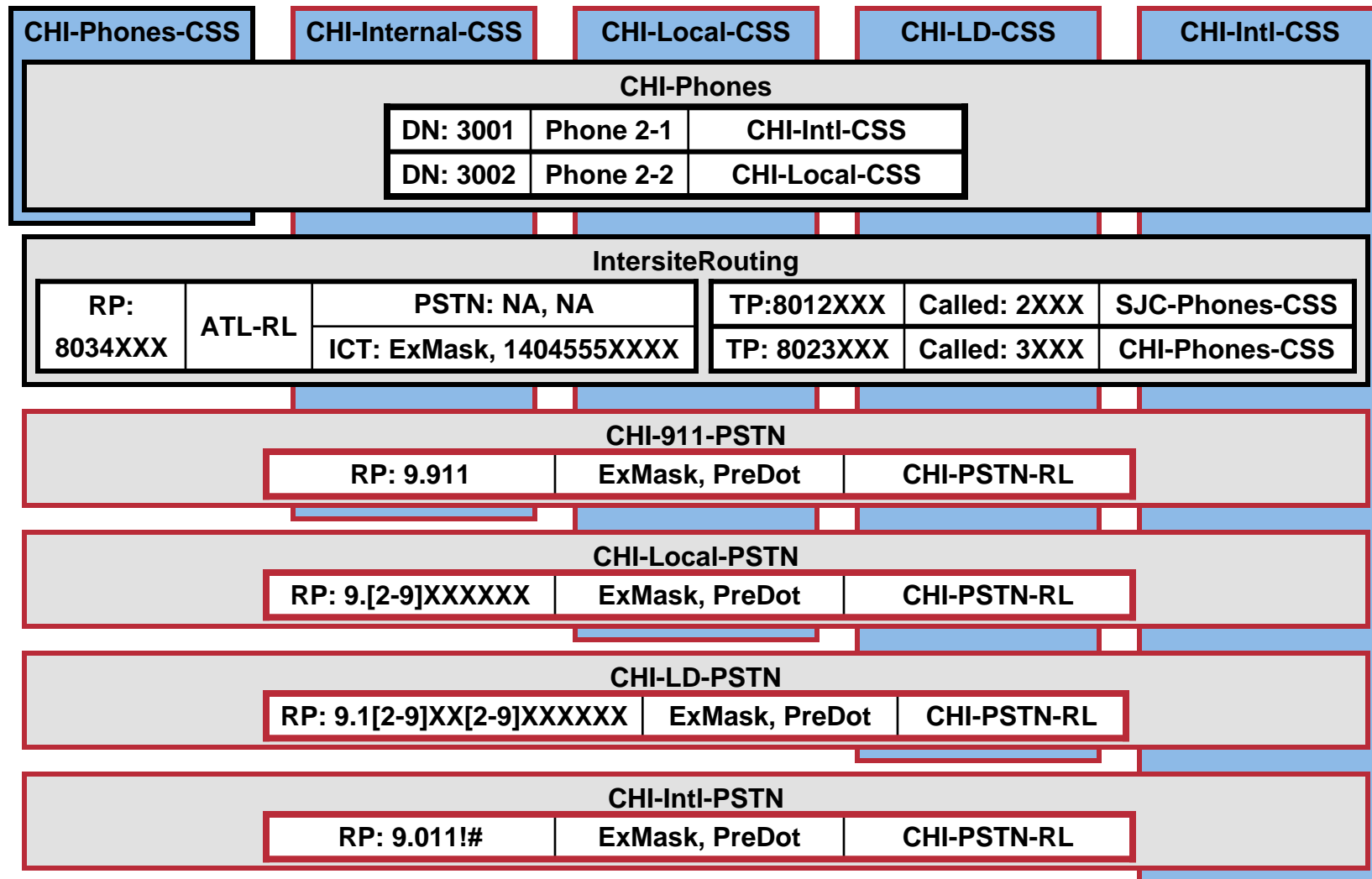
# Configuring PSTN Routing and Calling Privileges



# Configuring PSTN Routing and Calling Privileges: San Jose



# Configuring PSTN Routing and Calling Privileges: Chicago



# Configuring PSTN Routing and Calling Privileges: Revised Partitions and CSSs

**Find and List Partitions**

Status  
11 records found

Search Options  
Find partition where  begin   Search Within Results

Search Results

<input type="checkbox"/>	Partition Name	Description
<input type="checkbox"/>	<a href="#">CHI-911</a>	Chicago Emergency Calls
<input type="checkbox"/>	<a href="#">CHI-Intl-PSTN</a>	CHI Intl Routing
<input type="checkbox"/>	<a href="#">CHI-LD-PSTN</a>	CHI LD PSTN Routing
<input type="checkbox"/>	<a href="#">CHI-Local-PSTN</a>	CHI Local PSTN Routing
<input type="checkbox"/>	<a href="#">CHI-Phones</a>	Chicago Phones
<input type="checkbox"/>	<a href="#">Intersite-Routing</a>	Intersite Routing
<input type="checkbox"/>	<a href="#">SJC-911</a>	San Jose Emergency Calls
<input type="checkbox"/>	<a href="#">SJC-Intl-PSTN</a>	SJC Intl Routing
<input type="checkbox"/>	<a href="#">SJC-LD-PSTN</a>	SJC LD PSTN Routing
<input type="checkbox"/>	<a href="#">SJC-Local-PSTN</a>	SJC Local PSTN Routing
<input type="checkbox"/>	<a href="#">SJC-Phones</a>	SanJose Phones

Add New Select All Clear All Delete Selected Rows per Page 50

# Configuring PSTN Routing and Calling Privileges: Revised Partitions and CSSs

**Find and List Calling Search Spaces**

Status: 12 records found

Search Options: Find where Name begins with Find Search Within Results

Search Results:

	CSS Name	Description	Copy
<input type="checkbox"/>	<a href="#">CHI-Everywhere-CSS</a>		
<input type="checkbox"/>	<a href="#">CHI-Internal-CSS</a>		
<input type="checkbox"/>	<a href="#">CHI-Intl-CSS</a>		
<input type="checkbox"/>	<a href="#">CHI-LD-CSS</a>		
<input type="checkbox"/>	<a href="#">CHI-Local-CSS</a>		
<input type="checkbox"/>	<a href="#">CHI-Phones-CSS</a>		
<input type="checkbox"/>	<a href="#">SJC-Everywhere-CSS</a>		
<input type="checkbox"/>	<a href="#">SJC-Internal-CSS</a>		
<input type="checkbox"/>	<a href="#">SJC-Intl-CSS</a>		
<input type="checkbox"/>	<a href="#">SJC-LD-CSS</a>		
<input type="checkbox"/>	<a href="#">SJC-Local-CSS</a>		
<input type="checkbox"/>	<a href="#">SJC-Phones-CSS</a>		

Add New Select All Clear All Delete Selected Rows per Page 50

Done 192.168.1.10:8443

**Configure all CSSs.**

# Configuring PSTN Routing and Calling Privileges: Additional Route Lists

The screenshot shows the Cisco CallManager Administration interface in a Mozilla Firefox browser window. The browser title is "Find and List Route Lists - Mozilla Firefox" and the address bar shows the URL: `https://192.168.1.10:8443/ccmadmin/routeListTwoFindList.do?lookup=false&multiple=true&recCnt=0&colCnt=7`. The navigation bar indicates "Cisco CallManager Administration".

Route List	Route Groups	Gateway/Trunk	Digit Manipulation	Description
SJC-PSTN-RL	SJC-PSTN	Router1	NA	Route List for SJC PSTN Calls
CHI-PSTN-RL	CHI-PSTN	192.168.2.254	NA	Route List for CHI PSTN Calls

3 records found

**Search Options**  
Find Route List where  begins with    Search Within Results  
(device.name begins with any)

**Search Results**

	Name	Description	Enabled	Status
<input type="checkbox"/>	<a href="#">ATL-RL</a>		true	Registered with 192.168.1.10
<input type="checkbox"/>	<a href="#">CHI-PSTN-RL</a>		true	Registered with 192.168.1.10
<input type="checkbox"/>	<a href="#">SJC-PSTN-RL</a>		true	Registered with 192.168.1.10

Buttons: Add New, Select All, Clear All, Delete Selected, Reset Selected. Rows per Page: 50

Done 192.168.1.10:8443

**New RLs for PSTN routing.**

# Configuring PSTN Routing and Calling Privileges: Additional Route Patterns

Pattern	Partition	Type	Destination	Digit Manipulation	Description
<b>PSTN Routing</b>					
911 9.911	SJC-911-PSTN CHI-911-PSTN	RP RP	SJC-PSTN-RL CHI-PSTN-RL	Called: PreDot Calling: ExMask	Emergency Calls
9.[2-9]XXXXXX	SJC-Local-PSTN CHI-Local-PSTN	RP RP	SJC-PSTN-RL CHI-PSTN-RL	Called: PreDot Calling: ExMask	Local PSTN Routing
9.1[2-9]XX[2-9]XXXXXX	SJC-LD-PSTN CHI-LD-PSTN	RP RP	SJC-PSTN-RL CHI-PSTN-RL	Called: PreDot Calling: ExMask	LD PSTN Routing
9.011! 9.011!#	SJC-Intl-PSTN CHI-Intl-PSTN	RP RP	SJC-PSTN-RL CHI-PSTN-RL	Called: PreDot Calling: ExMask	Intl PSTN Routing

# Configuring PSTN Routing and Calling Privileges: Additional Route Patterns

The screenshot displays the Cisco CallManager Administration web interface in Mozilla Firefox. The page title is "Route Pattern Configuration". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The "Route Pattern Configuration" section shows the following details:

- Status:** Ready
- Pattern Definition:**
  - Route Pattern\*: 9.1[2-9]XX[2-9]XXXXXX
  - Route Partition: SJC-LD-PSTN
  - Description:
  - Numbering Plan\*: NANP
  - Route Filter: < None >
  - MLPP Precedence\*: Default
  - Gateway/Route List\*: SJC-PSTN-RL (Edit)
  - Route Option:
    - Route this pattern
    - Block this pattern
  - Call Classification\*: OffNet
  - Allow Device Override
  - Provide Outside Dial Tone
  - Allow Overlap Sending
  - Urgent Priority
  - Require Forced Authorization Code
  - Authorization Level\*: 0

Two yellow callout boxes highlight key configuration elements:

- Typical US LD Pattern.** Points to the "Route Pattern\*" field containing the pattern `9.1[2-9]XX[2-9]XXXXXX`.
- Select appropriate PSTN RL.** Points to the "Gateway/Route List\*" dropdown menu, which is currently set to "SJC-PSTN-RL".



# Configuring PSTN Routing and Calling Privileges: Additional Route Patterns

Route Pattern Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccmadmin/routePattern2Edit.do?key=1025a559-31d4-3393-fee2-df5529d50fad

Authorization Level: [ ]

Require Client Matter Code

**Calling Party Transformations**

Use Calling Party's External Phone Number Mask

Calling Party Transform Mask: [ ]

Prefix Digits (Outgoing Calls): [ ]

Calling Line ID Presentation\*: [ Default ]

Calling Name Presentation\*: [ Default ]

**Connected Party Transformations**

Connected Line ID Presentation\*: [ Default ]

Connected Name Presentation\*: [ Default ]

**Called Party Transformations**

Discard Digits: [ PreDot ]

Called Party Transform Mask: [ ]

Prefix Digits (Outgoing Calls): [ ]

**ISDN Network-Specific Facilities Information Element**

Network Service Protocol: [ - Not Selected - ]

Carrier Identification Code: [ ]

Network Service	Service Parameter Name	Service Parameter Value
[ - Not Selected - ]	[ < Not Exist > ]	[ ]

[ Save ] [ Delete ] [ Copy ] [ Add New ]

Done 192.168.1.10:8443

**Ensure 9 is stripped off using PreDot Discard.**

# Configuring PSTN Routing and Calling Privileges: Additional Route Patterns

The screenshot displays a web browser window titled "Route Plan Report - Mozilla Firefox". The address bar shows the URL: <https://192.168.1.10:8443/ccadmin/rprptFindList.do?recCnt=0>. The main content area is a table listing route patterns. A red rectangular box highlights the first six rows of the table. A yellow callout box with a speech bubble points to the highlighted area, containing the text: "San Jose PSTN RPs, RLs, RGs and GWs." The table has the following data:

Route Pattern	Route Pattern	Type	Resources
911	<a href="#">SJC-911</a>	Route Pattern	SJC-PSTN-RL SJC-PSTN MGCP_S1/DS1-0@Router1 (All ports)
9.911	<a href="#">SJC-911</a>	Route Pattern	SJC-PSTN-RL SJC-PSTN MGCP_S1/DS1-0@Router1 (All ports)
9.011 #	<a href="#">SJC-Intl-PSTN</a>	Route Pattern	SJC-PSTN-RL SJC-PSTN MGCP_S1/DS1-0@Router1 (All ports)
9.011!	<a href="#">SJC-Intl-PSTN</a>	Route Pattern	SJC-PSTN-RL SJC-PSTN MGCP_S1/DS1-0@Router1 (All ports)
9.1[2-9]XX[2-9]XXXXXX	<a href="#">SJC-LD-PSTN</a>	Route Pattern	SJC-PSTN-RL SJC-PSTN MGCP_S1/DS1-0@Router1 (All ports)
9.[2-9]XXXXXX	<a href="#">SJC-Local-PSTN</a>	Route Pattern	SJC-PSTN-RL SJC-PSTN MGCP_S1/DS1-0@Router1 (All ports)

At the bottom of the table, there is a "Rows per Page" dropdown menu set to "25". The browser status bar at the bottom shows "Done" and the IP address "192.168.1.10:8443".

# Configuring PSTN Routing and Calling Privileges: Additional Route Patterns

The screenshot shows a web browser window titled "Route Plan Report - Mozilla Firefox" with the URL <https://192.168.1.10:8443/ccadmin/rprptFindList.do?recCnt=0>. A search bar contains the text "Select item or enter search text". Below the search bar, the "Search Results" section displays a table with the following data:

Pattern/Directory Number	Partition	Type	Route Detail
<a href="#">9.911</a>	<a href="#">CHI-911</a>	Route Pattern	CHI-PSTN-RL CHI-PSTN H.323_192.168.2.254 (All ports)
<a href="#">911</a>	<a href="#">CHI-911</a>	Route Pattern	CHI-PSTN-RL CHI-PSTN H.323_192.168.2.254 (All ports)
<a href="#">9.011!</a>	<a href="#">CHI-Intl-PSTN</a>	Route Pattern	CHI-PSTN-RL CHI-PSTN H.323_192.168.2.254 (All ports)
<a href="#">9.011!#</a>	<a href="#">CHI-Intl-PSTN</a>	Route Pattern	CHI-PSTN-RL CHI-PSTN H.323_192.168.2.254 (All ports)
<a href="#">9.1[2-9]XX[2-9]XXXXXX</a>	<a href="#">CHI-LD-PSTN</a>	Route Pattern	CHI-PSTN-RL CHI-PSTN H.323_192.168.2.254 (All ports)
<a href="#">9.[2-9]XXXXXX</a>	<a href="#">CHI-Local-PSTN</a>	Route Pattern	CHI-PSTN-RL CHI-PSTN H.323_192.168.2.254 (All ports)

A yellow callout box on the right side of the screenshot contains the text: "Chicago PSTN RPs, RLs, RGs and GWs." A red rectangular box highlights the table content.

# Configuring PSTN Routing and Calling Privileges: Inbound PSTN CSS

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays the URL: `https://192.168.1.10:8443/ccadmin/gatewayEdit.do?key=e6522b89-6892-5e87-0c2d-465faa7509ca`. The page title is "Gateway Configuration - Mozilla Firefox". The navigation menu includes "System", "Call Routing", "Media Resources", "User Management", "Bulk Administration", and "Help". The "Gateway Configuration" link is highlighted, and a yellow callout box with a circled "1" points to it, containing the text "Access the Gateway Configuration." Below the navigation menu, the "Status" section shows "Status: Ready". The "Device Information" section displays the following details:

Product	Cisco MGCP T1 Port
Gateway	Router1
Device Protocol	Digital Access PRI
Registration	Registered with Cisco CallManager 192.168.1.10
IP Address	192.168.1.254
End-Point Name *	S1/DS1-0@Router1
Description	S1/DS1-0@Router1
Device Pool*	SanJose
Call Classification*	Use System Default
NetworkLocale	< None >
Media Resource Group List	< None >
Location*	SJC
AAR Group	< None >
Load Information	

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10:8443".

# Configuring PSTN Routing and Calling Privileges: Inbound PSTN CSS

Gateway Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccmadmin/gatewayEdit.do?key=e6522b89-6892-5e87-0c2d-465faa7509ca

Channel IE Type\* Use Number when 1B

PCM Type\* μ-law

Delay for first restart (1/8 sec ticks)\* 32

Delay between restarts (1/8 sec ticks)\* 4

Inhibit restarts at PRI initialization

Enable status poll

Unattended Port

**Call Routing Information - Inbound Calls**

Significant Digits\* All

Calling Search Space SJC-Phones-CSS

AAR Calling Search Space < None >

Prefix DN

**Call Routing Information - Outbound Calls**

Calling Party Presentation\* Default

Calling Party Selection\* Originator

Called party IE number type unknown\* Cisco CallManager

Calling party IE number type unknown\* Cisco CallManager

Called Numbering Plan\* Cisco CallManager

Calling Numbering Plan\* Cisco CallManager

Number of digits to strip\* 0

Caller ID DN

SMDI Base Port\* 0

https://192.168.1.10:8443/ccmadmin/routeGroupEdit.do?key=0108045f-8eb8-1442-05f6-b7c3c515d737 192.168.1.10:8443

Select the appropriate CSS for inbound calls and Save/Reset, 2

# Summary

- **Predefined dial-plan design will be used.**
- **Intrasite routing are configured using partitions and CSSs.**
- **On-Cluster intersite routing are configured using Translation Patterns.**
- **Off-Cluster intersite routing are configured using Route Patterns, Route Lists and Route Groups.**
- **Calling Privileges are configured via partitions, CSSs and Route Patterns.**

# CISCO SYSTEMS





## **Configuring CallManager for Multi Site Deployments**

# **Understanding Cisco Unified CallManager Codecs and Call Admission Control**



# Objectives

- **Multi-Site Characteristics**
- **Cisco Unified CallManager 5.0 Supported Codecs**
- **Understanding Regions**
- **CAC Mechanisms**
- **Understanding Locations**
- **Understanding Resource Reservation Protocol (RSVP) Based CAC**

# Multi-Site Characteristics

## **Multi-Site deployments have the following characteristics:**

- **Calls travel across the WAN.**
  - **Bandwidth is more limited.**
  - **Quality is harder to control.**
- **Different codec might be used to save bandwidth.**
- **Number of concurrent calls needs to be controlled to reflect available bandwidth.**
  - **Oversubscribing available bandwidth leads to degraded voice quality for ALL active calls.**

# Cisco Unified CallManager 5.0 Supported Codecs

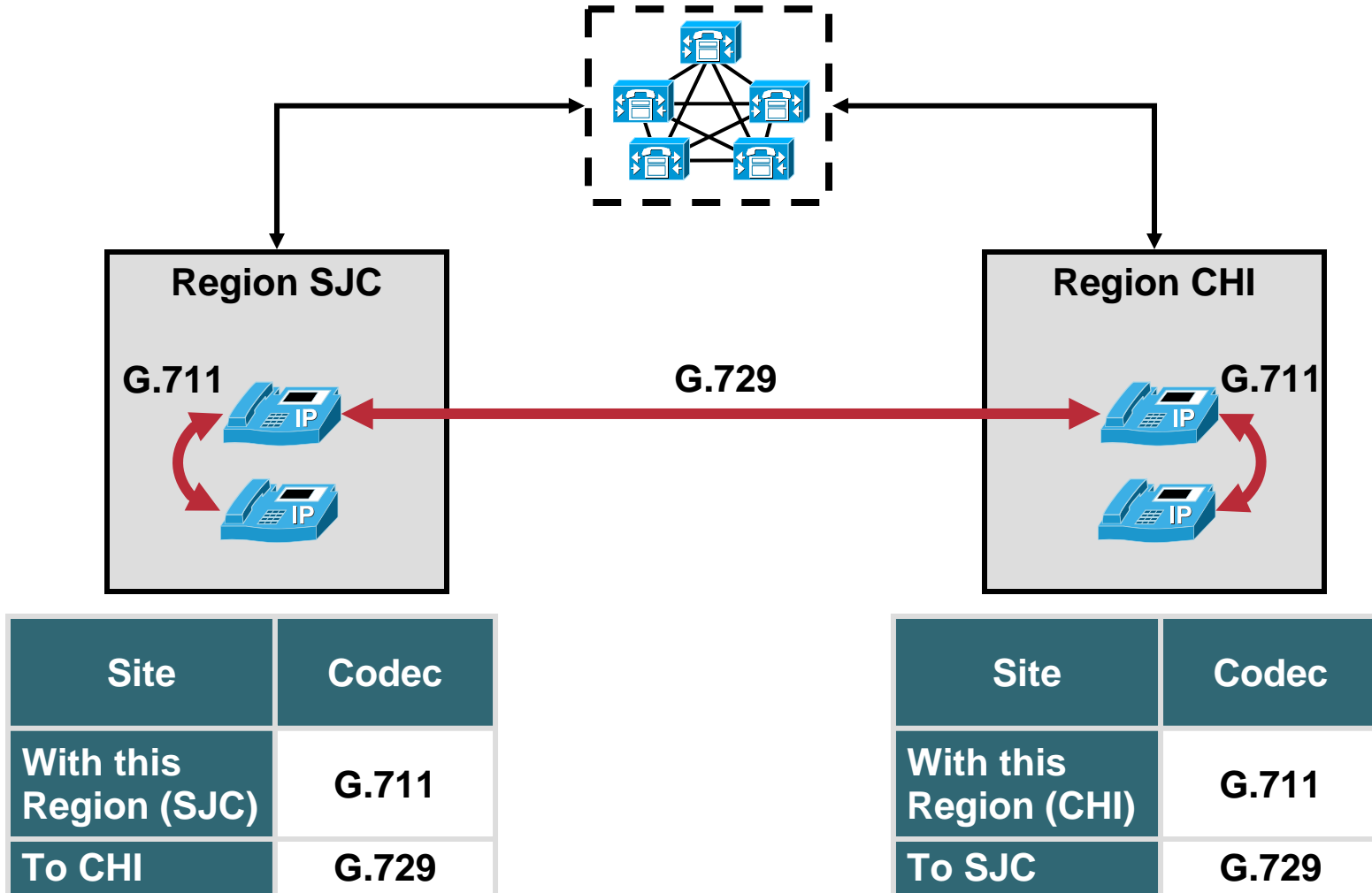
Codec	Layer 3 Bandwidth	Description
<b>G.711</b>	80 kbps	This codec represents the default codec for all Cisco Unified CallManager calls.
<b>G.722</b>	80 kbps	Video endpoints typically prefer this codec.
<b>G.723</b>	24 kbps	The system supports this low-bit-rate codec for use with older Cisco IP Phone model 12 SP Series and Cisco IP Phone model 30 VIP.
<b>G.728</b>	16 kbps	Video endpoints support this low-bit-rate codec.
<b>G.729</b>	24 kbps	The system supports this low bit-rate codec for Cisco IP Phone 7900 models.
<b>Wideband</b>	272 kbps	The system supports this high-quality, high-bandwidth audio codec for IP-phone to IP-phone calls that the Cisco IP Phone 7900 models support.
<b>GSM</b>	29 kbps	The Global System for Mobile Communications (GSM) codec enables the MNET system for GSM wireless handsets to interoperate with Cisco Unified CallManager.

# Understanding Regions

**CallManager uses regions to determine the codec to be used:**

- **Used for on-cluster codec selection.**
- **Regions are assigned to device pools.**
- **A matrix is used to define:**
  - **Codec for intra-region calls.**
  - **Codec for inter-region calls.**

# Understanding Regions

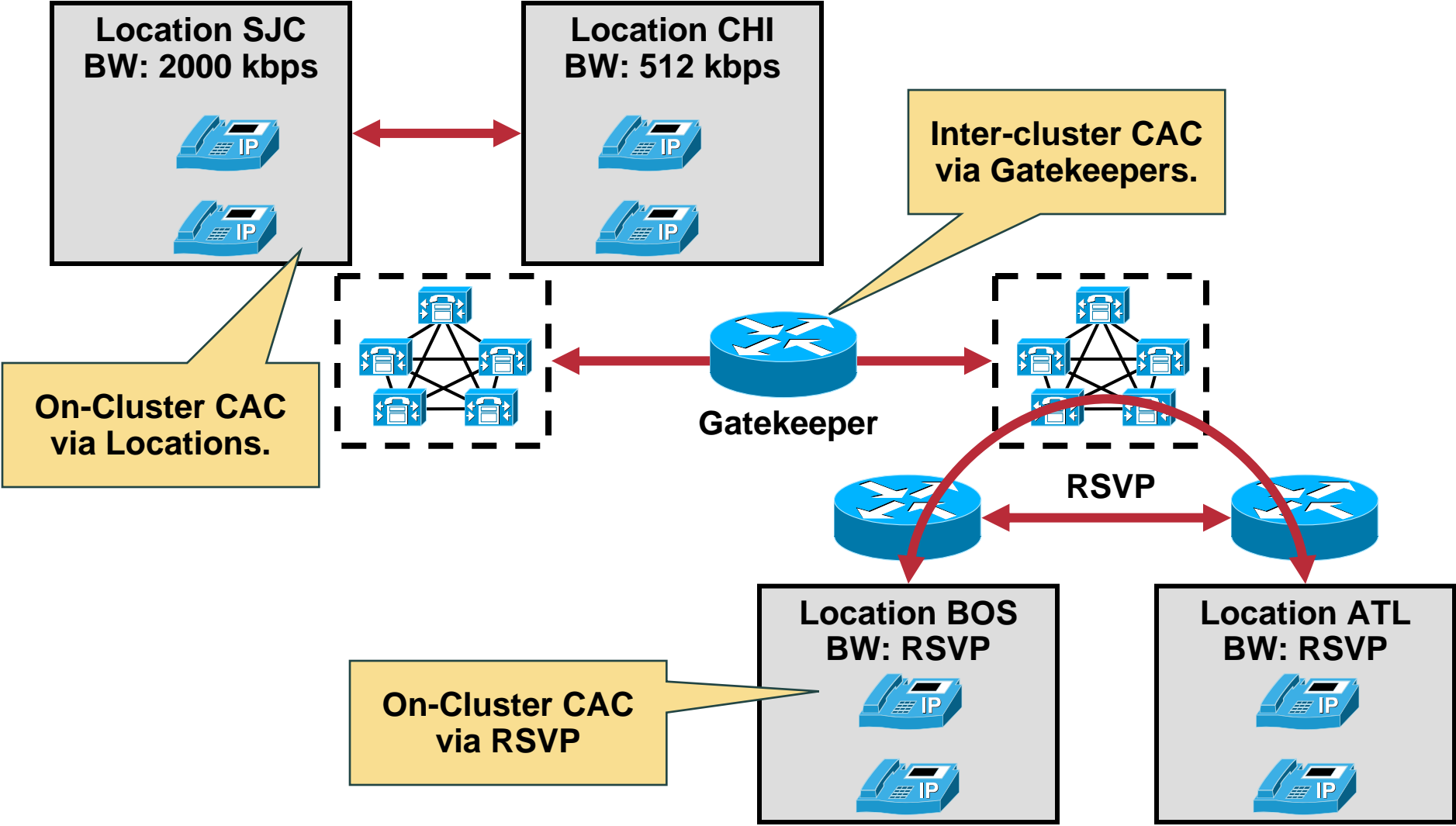


# CAC Mechanisms

**Cisco Unified CallManager 5.0 can use the following CAC mechanisms:**

- **On-Cluster CAC:**
  - **Location based CAC**
    - **Configured purely in Cisco Unified CallManager, Infrastructure unaware**
  - **RSVP based CAC**
    - **Uses Locations and IOS RSVP agents, Infrastructure aware**
- **Gatekeeper controlled**
  - **Used for inter-cluster routing**
  - **Bandwidth configured in gatekeepers, Infrastructure unaware**

# CAC Mechanisms



# Understanding Locations

## **Locations are configured in Cisco Unified CallManager:**

- **Devices are assigned to locations.**
- **Locations can control bandwidth individually for:**
  - **Audio**
  - **Video**
- **Bandwidth is statically defined by the administrator.**
  - **Option to use RSVP instead of bandwidth.**

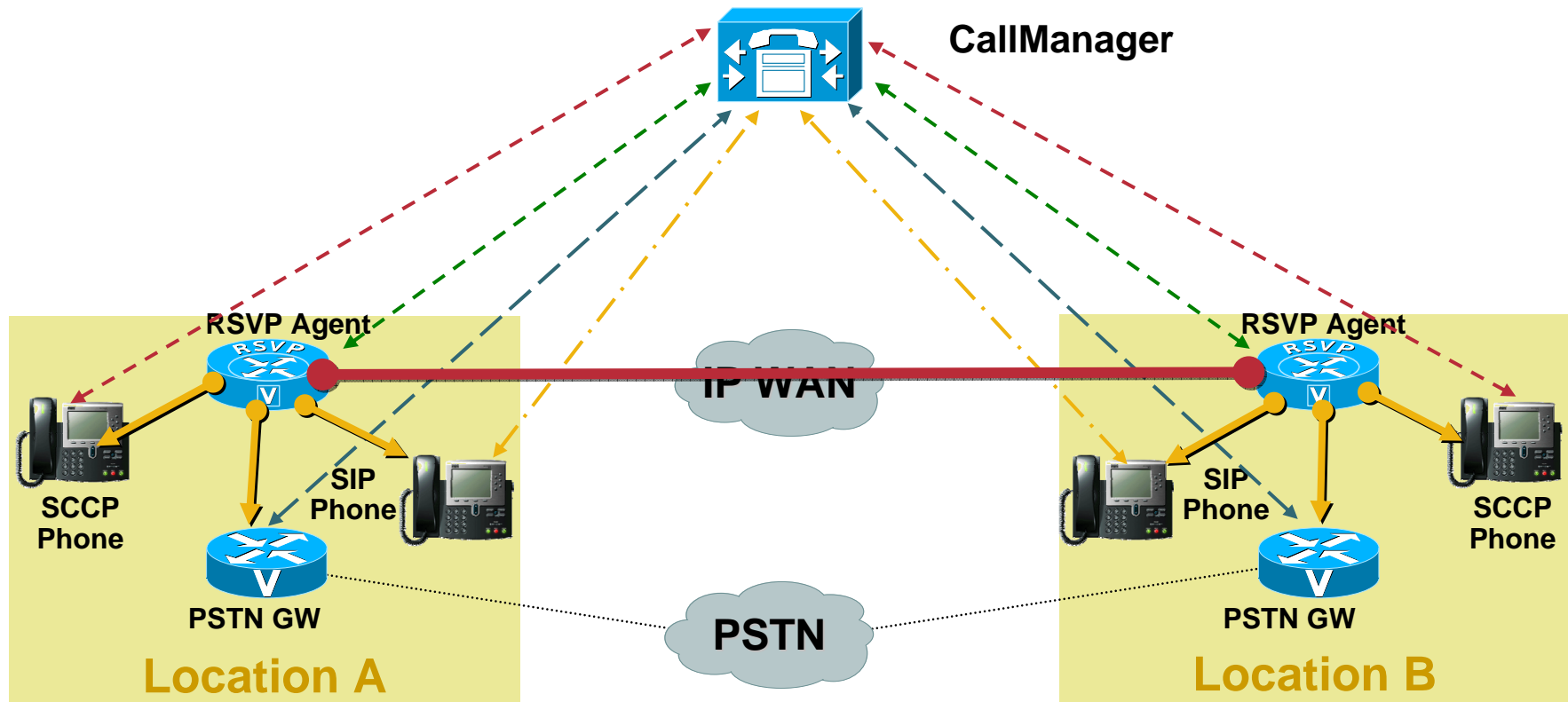


# Understanding RSVP-Based CAC

**RSVP can be used to control inter-location bandwidth:**

- **Configure locations to use RSVP.**
- **For every inter-location call:**
  - **Cisco Unified CallManager routes call using IOS Media Termination Point (MTP).**
  - **IOS gateways use RSVP to control MTP – MTP connections.**
- **Flexible, topology aware CAC.**

# Understanding RSVP-Based CAC



← - - - → **SCCP**

← - - - → **Media Resource Control**

← - - - → **MGCP or H.323**

—●—●— **RSVP-RTP**

—●—●— **RTP**

# Summary

- **Multi-Site deployments use various codecs and have the requirement to control bandwidth.**
- **Cisco Unified CallManager support various codecs.**
- **Regions are used for codec selections.**
- **Cisco Unified CallManager has three CAC mechanisms: Locations, Locations with RSVP and GK controlled CAC.**
- **Location based CAC is simply to set-up but topology unaware.**
- **RSVP can extend location based CAC via IOS MTPs using RSVP for bandwidth control.**

# CISCO SYSTEMS





## **Configuring CallManager for Multi Site Deployments**

# **Configuring Regions, Locations and RSVP-Based CAC**

# Objectives

- **Regions Configuration**
- **Locations Configuration**
- **RSVP Concepts**
- **RSVP-Based CAC Configuration on Cisco CallManager**
- **RSVP-Based CAC Configuration on Gateways**

# Regions Configuration

## Region configuration steps:

1. Add regions and specify codecs.
2. Assign region to device pools.

# Regions Configuration

The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser. The browser's address bar shows the URL `https://192.168.1.10:8443/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The navigation menu on the left side of the page is expanded, showing the following items:

- Server
- Cisco CallManager
- Cisco CallManager Group
- Phone NTP Reference
- Date/Time Group
- Presence Group
- Region** (highlighted)
- Device Pool
- DHCP
- LDAP
- Location
- SRST
- MLPP Domain
- Enterprise Parameters
- Service Parameters
- Security Profile
- Application Server
- Licensing

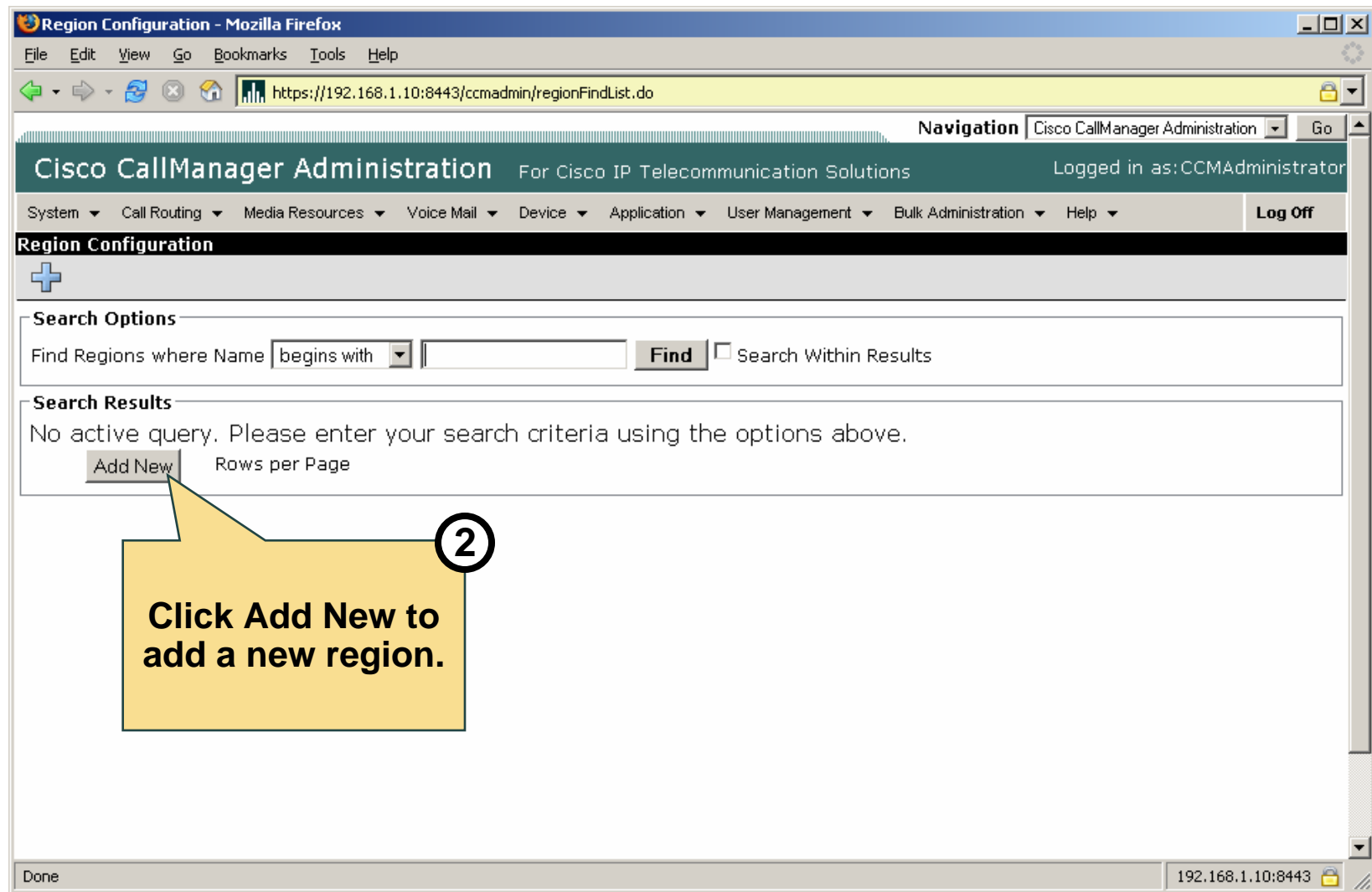
A yellow callout box with a black border contains the text "Go to System > Region." with an arrow pointing to the "Region" menu item. A circled number "1" is placed next to the "Region" menu item.

The main content area of the page displays "Cisco CallManager Administration" and some system information, including "1.51-414" and "n: 1.1.0.0-1".

The browser's status bar at the bottom shows the URL `https://192.168.1.10:8443/ccmadmin/regionFindList.do` and the IP address `192.168.1.10:8443`.



# Regions Configuration



The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/regionFindList.do`. The page title is "Region Configuration" and it includes a navigation menu with options like "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". A "Log Off" button is also present. The main content area features a search section with a dropdown menu set to "begins with" and a "Find" button. Below the search section, there is a "Search Results" area with the message "No active query. Please enter your search criteria using the options above." and an "Add New" button. A yellow callout box with a circled "2" points to the "Add New" button, containing the text "Click Add New to add a new region." The status bar at the bottom shows "Done" and the IP address "192.168.1.10:8443".

# Regions Configuration

Region Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccmadmin/regionEdit.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as:CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Region Configuration Related Links: Back To Find/List Go

Region Information

Name\* SJC

Save

i \*- indicates r

3 Enter a name for the new region and save.

Done 192.168.1.10:8443

# Regions Configuration

The screenshot shows the Cisco CallManager Administration interface in Mozilla Firefox. The page title is "Region Configuration - Mozilla Firefox" and the URL is "https://192.168.1.10:8443/ccmadmin/regionSave.do". The navigation bar includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The main menu has "System", "Call Management", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Region Configuration" page has a "Save" button (callout 3) and a "Status" section with "Add successful". The "Region Information" section has a "Name\*" field with "SJC". The "Region Relationships" section has a table with columns "Region", "Audio Codec", and "Video Call Bandwidth". The "Modify Relationship to other Regions" section has a "Regions" list with "Default" and "SJC" (callout 1), and an "Audio Codec" dropdown with "G.729" (callout 2).

**3** Save the configuration.

**1** Select a region from the list.

**2** Select the codec to use with the previously selected region.

# Regions Configuration

The screenshot shows the Cisco CallManager Administration interface for Region Configuration. The page title is "Region Configuration" and the user is logged in as "CCMAdministrator". The navigation menu includes System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "Region Configuration" section shows a "Status" of "Update successful" and "Region Information" with "Name\*" set to "SJC".

The "Region Relationships" table is highlighted with a red border and contains the following data:

Region	Audio Codec	Bandwidth
CHI	G.729	384
Default	G.729	384
SJC	G.711	384

Two callouts provide additional context:

- Use G.729 with remote regions.** (Pointing to the G.729 entries for CHI and Default)
- Use G.711 within regions SJC (intrasite calls).** (Pointing to the G.711 entry for SJC)

The "Modify Relationship to other Regions" section shows a list of regions (CHI, Default, SJC) and a dropdown menu currently set to "G.729". Radio buttons for "Keep Current Setting" and "Use System Default" are also visible.

# Regions Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The page title is "Device Pool Configuration". The navigation menu includes "Cisco CallManager Administration" and "Go". The user is logged in as "CCMAdministrator". The "Device Pool Configuration" section is active, showing the "SanJose" device pool with 11 members. The "Device Pool Settings" section is expanded, showing various configuration options. The "Region" dropdown menu is highlighted with a red box, and a yellow callout bubble points to it with the text "Assign region to device pool and save/reset.".

Field	Value
Device Pool Name*	SanJose
Cisco CallManager Group*	Default
Date/Time Group*	CMLocal
Region*	SJC
Softkey Template*	Standard User
SRST Reference*	Use Default Gateway
Calling Search Space for Auto-registration	< None >
Media Resource Group List	< None >
Network Hold MOH Audio Source	< None >
User Hold MOH Audio Source	< None >
Network Locale	< None >
User Locale	< None >

# Locations Configuration

## Locations configuration steps:

1. Add locations and configure bandwidth.
2. Assign location to device.

# Locations Configuration

Cisco CallManager Console - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccmadmin/showHome.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

- Server
- Cisco CallManager
- Cisco CallManager Group
- Phone NTP Reference
- Date/Time Group
- Presence Group
- Region
- Device Pool
- DHCP
- LDAP
- Location**
- SRST
- MLPP Domain
- Enterprise Parameters
- Service Parameters
- Security Profile
- Application Server
- Licensing

**Go to System > Location.**

Cisco CallManager Administration

System version: 5.0.1.51-414  
Administration version: 1.1.0.0-1

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ates and local country laws governing import, export, transfer and use. Delivery of Cisco  
export, distribute or use encryption. Importers, exporters, distributors and users are  
is product you agree to comply with applicable laws and regulations. If you are unable to

e found at: <http://www.cisco.com/wwl/export/crypto/tool/starg.html>.  
export@cisco.com.

https://192.168.1.10:8443/ccmadmin/locationFindList.do 192.168.1.10:8443

# Locations Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/locationFindList.do`. The page title is "Find and List Locations". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The main content area is titled "Find and List Locations" and features a plus sign icon. Below this, there is a "Status" section indicating "0 records found". The "Search Options" section includes a search form with the text "Find Find and List Locations where", a dropdown menu set to "Location", a "begins with" dropdown, an empty text input field, a "Find" button, and a checkbox for "Search Within Results".

The "Search Results" section displays the message "No active query. Please enter your search criteria using the options above." Below this message is an "Add New" button and a "Rows per Page" label. A yellow callout box with a circled "2" points to the "Add New" button, containing the text: "Click Add New to add a new location."



# Locations Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows `https://192.168.1.10:8443/ccmadmin/locationEdit.do`. The page title is "Cisco CallManager Administration" and the user is logged in as "CCMAdministrator". The navigation menu includes System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "Location Configuration" section is active, showing a "Status: Ready" and "Location Information" section with a "Name" field containing "SJC". The "Audio Calls Information" section has "Audio Bandwidth" set to "2000" kbps, and the "Video Calls Information" section has "Video Bandwidth" set to "384". The "Modify Setting(s) to Other Locations" section shows a list of locations with "Hub\_None" selected. Three callout boxes provide instructions: 1. "Specify location name." points to the "Name" field. 2. "Specify bandwidth for location or select unlimited." points to the "Audio Bandwidth" field. 3. "Only important for RSVP based CAC." points to the "Hub\_None" selection in the "Modify Setting(s) to Other Locations" section.

Location Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Location Configuration Related Links: Back To Find/List Go

Status

Status: Ready

Location Information

Name SJC

Audio Calls Information

Audio Bandwidth \*  Unlimited  2000

If the audio quality is poor or choppy, lower the bandwidth to 56 kbps or 64 kbps.

Video Calls Information

Video Bandwidth \*  None  Unlimited  384

Modify Setting(s) to Other Locations

Location

Hub\_None

RSVP Setting

Use System Default

Done 192.168.1.10:8443

3 Specify location name.

4 Specify bandwidth for location or select unlimited.

Only important for RSVP based CAC.

# Locations Configuration

**Phone Configuration - Mozilla Firefox**

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccadmin/phoneEdit.do?key=a150cd93-c8b4-45ad-a9f2-334d5ca10293

**Phone Configuration** Related Links: Back To Find/List

Status: Ready

**Association Information**

Modify Button Items

- Line [1] - 4000 (no partition)
- Line [2] - Add a new DN
- Add a new SD
- Add a new SD
- Add a new SD
- Add a new SD
- Unassigned Associated Items -----
- Add a new SD
- Add a new SURL
- Add a new BLF SD
- Privacy
- None

**Phone Type**

Product Type: Cisco 7960  
Device Protocol: SCCP

**Device Information**

Registration	Unregistered
IP Address	10.128.128.76
MAC Address*	005345000000
Description	Phone 1-1
Device Pool*	SanJose
Phone Button Template*	Standard 7960 SCCP
Softkey Template	< None >
Common Phone Profile*	Standard Common P
Calling Search Space	< None >
AAR Calling Search Space	< None >
Media Resource Group List	< None >
User Hold Audio Source	< None >
Network Hold Audio Source	< None >
Location*	SJC

Assign device to correct location and save/reset.

Done 192.168.1.10:8443

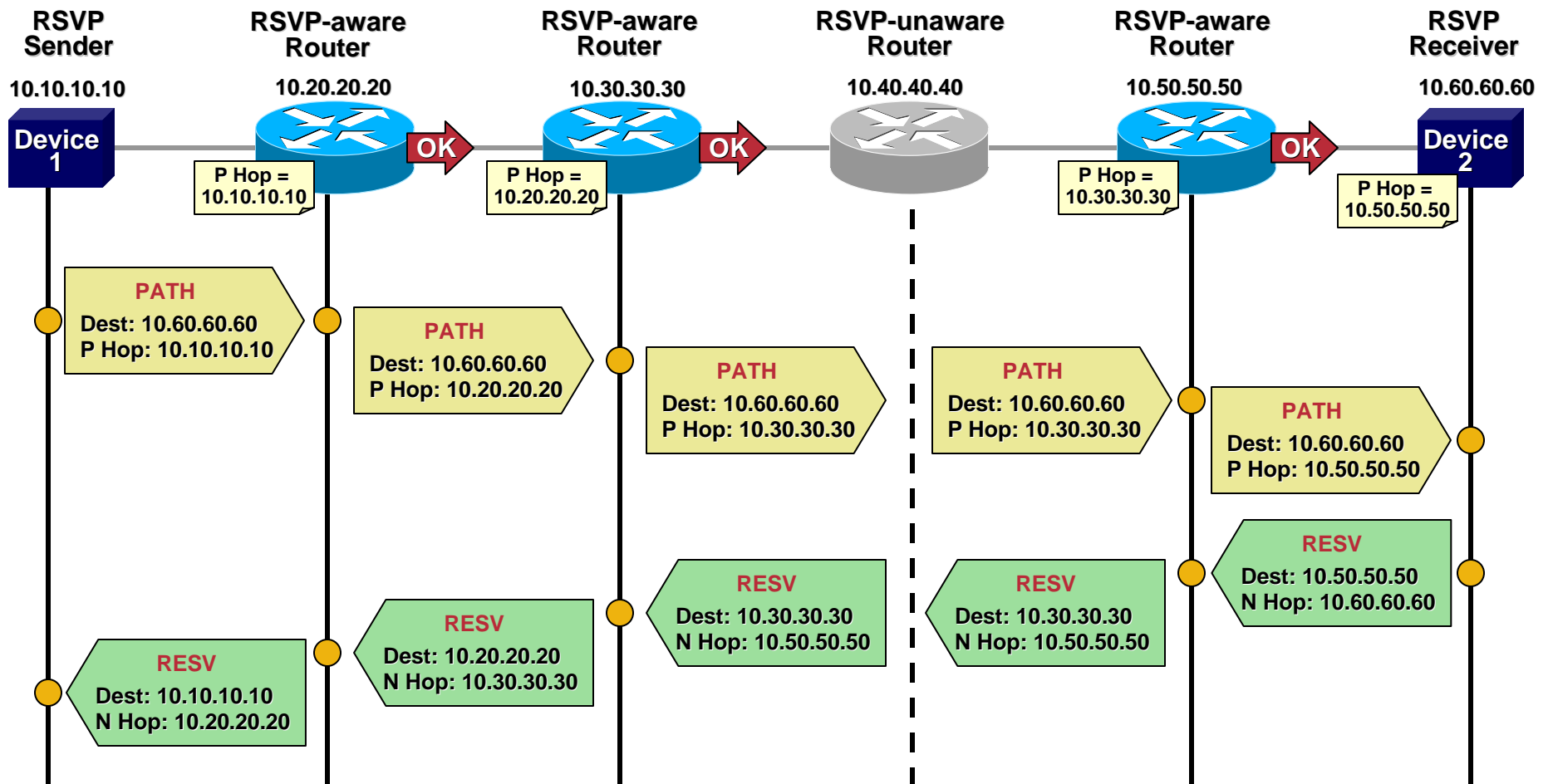
# RSVP Concepts

## **RSVP is used to reserve bandwidth:**

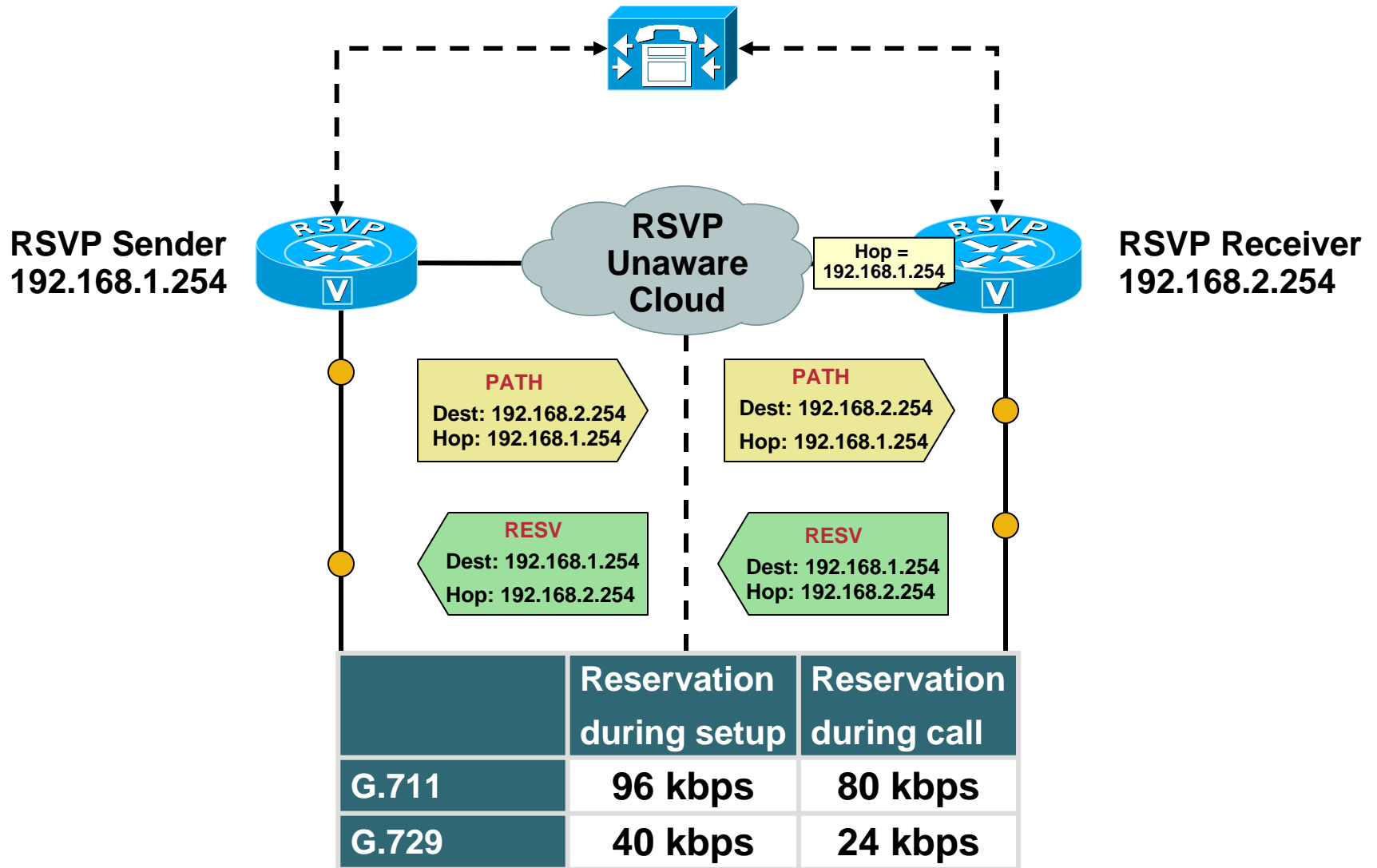
- **Bandwidth is reserved hop-by-hop.**
- **If all RSVP aware entities in a path reserve bandwidth:**
  - **Traffic can flow.**
- **RSVP unaware entities pass on RSVP packets.**
- **Minimum number of RSVP aware devices: two**

# RSVP Concepts

**Legend:** ● = RSVP processing occurs || ➔ OK = Bandwidth reserved on interface



# RSVP Concepts

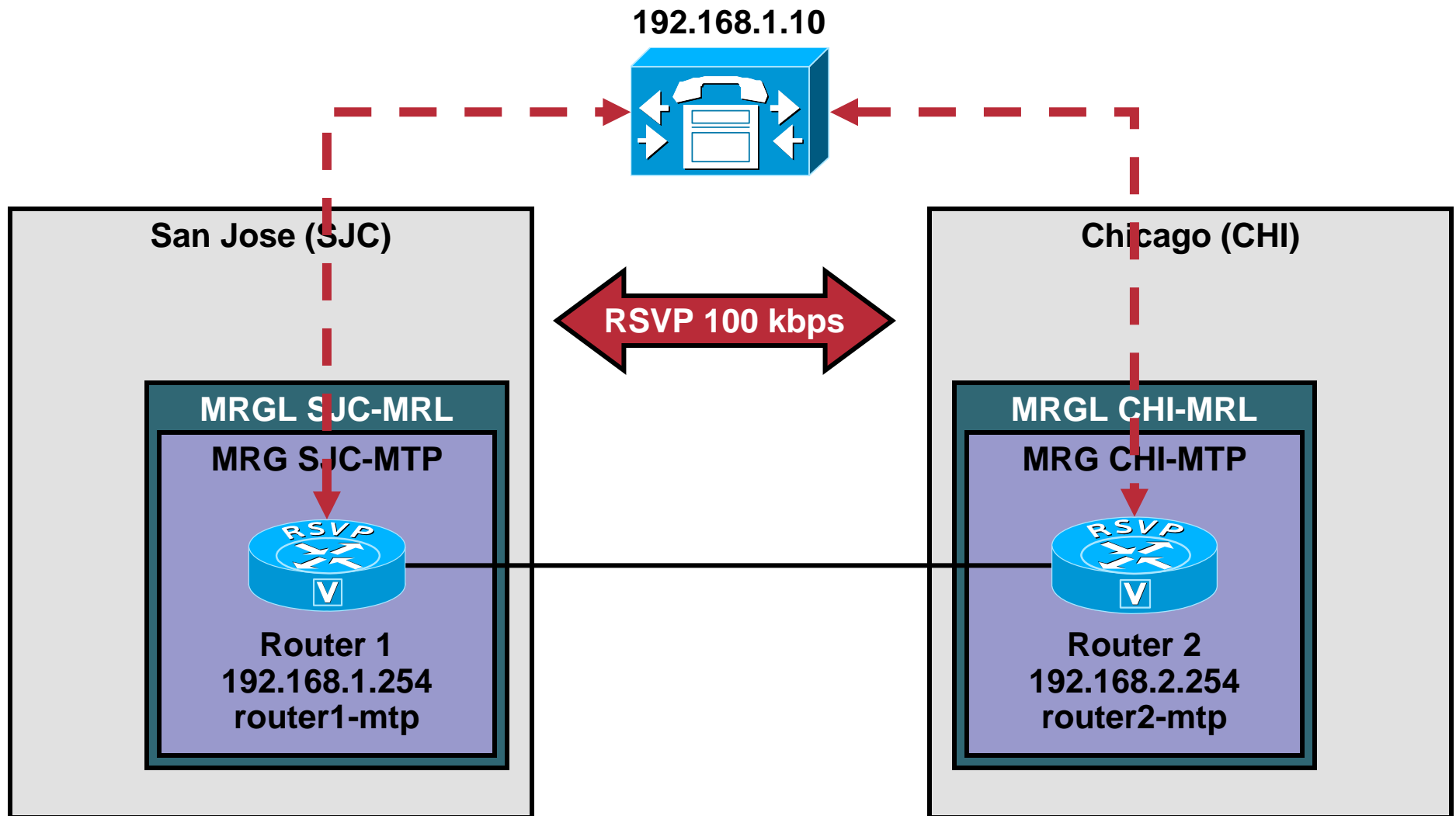


# RSVP-Based CAC Configuration Steps on Cisco Unified CallManager

## RSVP based CAC configuration steps on CallManager:

1. Add IOS MTP.
2. Add a Media Resource Group including the MTP.
3. Add/Modify a Media Resource Group List.
4. Assign Media Resource Group List to device or device pool.
5. Enable RSVP for specific locations.

# RSVP-Based CAC Configuration Steps on Cisco Unified CallManager



# Media Termination Point Configuration

The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser. The address bar shows the URL `https://192.168.1.10:8443/ccmadmin/showHome.do`. The page title is "Cisco CallManager Administration" and it is logged in as "CCMAdministrator". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", and "Device". The "Media Resources" dropdown menu is open, listing various options: "Annunciator", "Conference Bridge", "Media Termination Point" (highlighted), "Music On Hold Audio Source", "Fixed MOH Audio Source", "Music On Hold Server", "Transcoder", "Media Resource Group", "Media Resource Group List", and "MOH Audio File Management". A yellow callout box with a circled "1" points to the "Media Termination Point" option, containing the text "Go to Media Ressources > Media Termination Point." Below the menu, the page content includes the title "Cisco CallManager Administration", system version "5.0.1.51-414", administration version "1.1.0.0-1", and copyright information "© 1999 - 2005 Cisco Systems, Inc. All rights reserved." At the bottom of the browser window, the address bar shows `https://192.168.1.10:8443/ccmadmin/mtpFindList.do` and the IP address `192.168.1.10:8443`.

This product contains cryptographic features and is subject to United States and local country laws governing import, export, transfer and use. Delivery of Cisco cryptographic products does not imply third-party authority to import, export, distribute or use encryption. Importers, exporters, distributors and users are responsible for compliance with U.S. and local country laws. By using this product you agree to comply with applicable laws and regulations. If you are unable to comply with U.S. and local laws, return this product immediately.

A summary of U.S. laws governing Cisco cryptographic products may be found at: <http://www.cisco.com/wwl/export/crypto/tool/starg.html>. If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).



# Media Termination Point Configuration

Find and List Media Termination Points - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccmadmin/mtpFindList.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as:CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Find and List Media Termination Points

+ Add New

SQLException: Column (null) not found in any table in the query (or SLV is undefined).

Name begins with Find Search Within Results

Search

No active query. Please enter your search criteria using the options above.

Add New Select All Clear All Delete Selected Reset Selected

Done 192.168.1.10:8443

# Media Termination Point Configuration

The screenshot displays the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10:8443/ccmadmin/mtpEdit.do`. The page title is "Media Termination Point Configuration". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The main configuration area is titled "Media Termination Point Configuration" and includes the following fields:

- Status:** Status: Ready
- Media Termination Point Information:**
  - Media Termination Point Type\*: Cisco IOS Enhanced Software Media Termination Point
  - Media Termination Point Name\*: router1-mtp
  - Description: (empty)
  - Device Pool\*: SanJose

At the bottom of the form are "Save" and "Reset" buttons. A legend indicates that asterisks (\*) denote required fields.

Numbered callouts provide the following instructions:

- 3: Select Cisco IOS Enhanced Software MTP.
- 4: Specify MTP name.
- 5: Assign Device Pool.
- 6: Save configuration.

# Media Resource Group and List Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccmadmin/mrsrcGroupFindList.do`. The page title is "Find and List Media Resource Groups". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Media Resources" dropdown menu is open, showing options: "Annunciator", "Conference Bridge", "Media Termination Point", "Music On Hold Audio Source", "Fixed MOH Audio Source", "Music On Hold Server", "Transcoder", "Media Resource Group", "Media Resource Group List", and "MOH Audio File Management". The "Media Resource Group" option is highlighted. A yellow callout box with a circled "1" points to this option, containing the text: "Go to Media Resources > Media Resource Group." The main content area shows a search form with a "Find" button and a "Search Within Results" checkbox. The status bar at the bottom shows the URL and IP address `192.168.1.10:8443`.

# Media Resource Group and List Configuration

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10:8443/ccmadmin/mrsrcGroupEdit.do`. The page title is "Media Resource Group Configuration - Mozilla Firefox". The navigation menu includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The main content area is titled "Media Resource Group Configuration" and contains several sections:

- Status:** Shows "Status: Ready".
- Media Resource Group:** A dropdown menu set to "New".
- Media Resource Group Information:** Fields for "Name \*" (containing "SJC-MTP") and "Description" (containing "SJC-MTP").
- Devices for this Group:** Two lists: "Available Media Resources \*\*" (containing ANN\_2 (ANN), CFB\_2 (CFB), MOH\_2 (MOH), MTP\_2 (MTP), router2-mtp (MTP)) and "Selected Media Resources \*" (containing router1-mtp (MTP)).

Four numbered callouts provide instructions:

- Callout 1: "Save configuration." points to the save icon.
- Callout 2: "Specify name and description." points to the Name and Description fields.
- Callout 3: "Add previously configured MTP to MRG and save." points to the Selected Media Resources list.
- Callout 4: "Save configuration." points to the save icon.

# Media Resource Group and List Configuration

The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser. The browser's address bar displays the URL `https://192.168.1.10:8443/ccmadmin/showHome.do`. The page header includes the title "Cisco CallManager Administration" and the user "Logged in as: CCMAdministrator". A navigation menu at the top contains several categories: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. The "Media Resources" menu is expanded, showing a list of options: Annunciator, Conference Bridge, Media Termination Point, Music On Hold Audio Source, Fixed MOH Audio Source, Music On Hold Server, Transcoder, Media Resource Group, Media Resource Group List (highlighted), and MOH Audio File Management. A yellow callout box with a circled "1" points to the "Media Resource Group List" option and contains the text "Go to Media Resources > Media Resource List." Below the navigation menu, there is a copyright notice and a link to Cisco's export policy page.

Go to Media Resources > Media Resource List.

This product contains cryptographic features and is subject to United States and local country laws governing import, export, transfer and use. Delivery of Cisco cryptographic products does not imply third-party authority to import, export, distribute or use encryption. Importers, exporters, distributors and users are responsible for compliance with U.S. and local country laws. By using this product you agree to comply with applicable laws and regulations. If you are unable to comply with U.S. and local laws, return this product immediately.

A summary of U.S. laws governing Cisco cryptographic products may be found at: <http://www.cisco.com/wwl/export/crypto/tool/starg.html>.  
If you require further assistance please contact us by sending email to [export@cisco.com](mailto:export@cisco.com).

`https://192.168.1.10:8443/ccmadmin/mrsrclstFindList.do` 192.168.1.10:8443

# Media Resource Group and List Configuration

Media Resource Group List Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccadmin/mrsrListEdit.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as:CCMAdministrator

System Call Routing Media Resources Voice Mail Application User Management Bulk Administration Help Log Off

Media Resource Group List Configuration Related Links: Back To Find/List Go

Media Resource Group List Information

Name\* SJC-MRL

Media Resource Groups for this List

Available Media Resource Groups CHI-MTP

Selected Media Resource Groups SJC-MTP SJC-MOH SJC-CFB SJC-ANN

Save

\*- indicates required

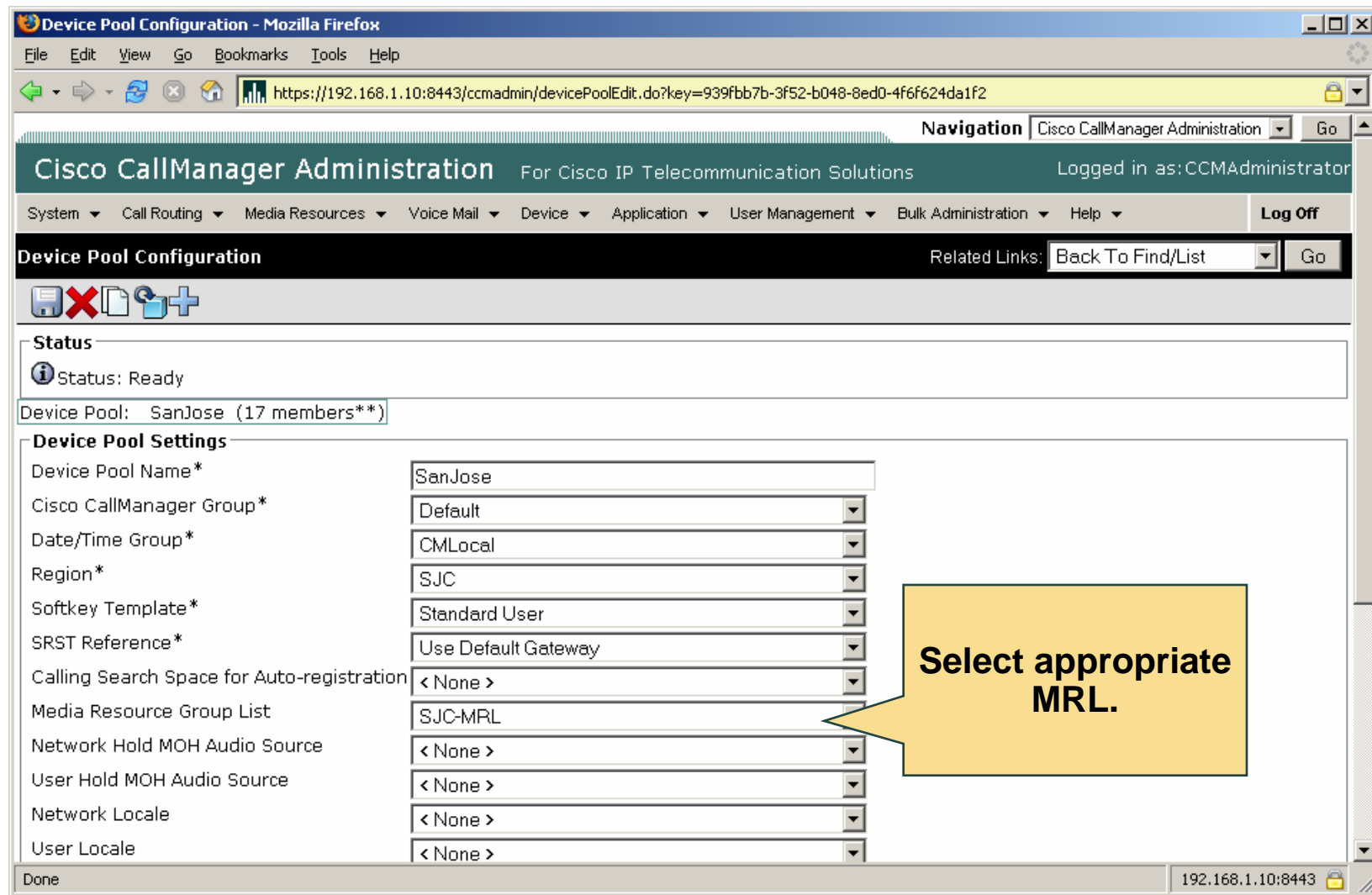
Read 192.168.1.10 192.168.1.10:8443

2 Specify name for MRGL.

3 Make sure to include the MRG which contains the MTP.

4 Save configuration.

# Media Resource Group and List Configuration



The screenshot displays the Cisco CallManager Administration web interface in Mozilla Firefox. The page title is "Device Pool Configuration" and the user is logged in as "CCMAdministrator". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Device Pool Configuration" section shows the "SanJose" device pool with 17 members. The "Device Pool Settings" table is as follows:

Setting	Value
Device Pool Name*	SanJose
Cisco CallManager Group*	Default
Date/Time Group*	CMLocal
Region*	SJC
Softkey Template*	Standard User
SRST Reference*	Use Default Gateway
Calling Search Space for Auto-registration	< None >
Media Resource Group List	SJC-MRL
Network Hold MOH Audio Source	< None >
User Hold MOH Audio Source	< None >
Network Locale	< None >
User Locale	< None >

A yellow callout box with the text "Select appropriate MRL." points to the "Media Resource Group List" dropdown menu.

# RSVP-Based CAC Configuration on Cisco Unified CallManager

The screenshot displays the 'Location Configuration' page in Mozilla Firefox. The browser address bar shows the URL: <https://192.168.1.10:8443/ccmadmin/locationEdit.do?key=8c23696d-998f-5ddb-d40b-dd0e3b357a54>. The page content includes:

- Status:** Status: Ready
- Location Information:** Name\* SJC
- Audio Calls Information:** Audio Bandwidth \*  Unlimited  2000 kbps. Note: If the audio quality is poor or choppy, lower the bandwidth setting. For ISDN, use multiples of 56 kbps or 64 kbps.
- Video Calls Information:** Video Bandwidth \*  None  Unlimited  384 kbps
- Location RSVP Settings:** Location SJC, RSVP Setting No Reservation. NOTE: Location(s) not displayed.
- Modify Setting(s) to Other Locations:** Location dropdown menu with options: CHI, Hub\_None, SJC. A callout '1' points to this menu with the text 'Select Location for RSVP.'
- Mandatory:** A dropdown menu with the option 'Mandatory' selected. A callout '2' points to this menu with the text 'Select Mandatory to activate RSVP.'

At the bottom of the page, there are buttons for 'Save', 'Delete', 'Copy', 'Add New', and 'Resync Bandwidth'. The status bar at the bottom left shows 'Done' and the bottom right shows the IP address '192.168.1.10:8443'.



# RSVP-Based CAC Configuration on Gateways

## RSVP CAC IOS configuration steps:

1. Specify CallManager.
2. Create SCCP group and point to profile.
3. Configure DSP Farm profile for software MTP.
4. Allocate IP RSVP bandwidth to interface(s)

# RSVP-Based CAC Configuration on Gateways

```
!  
sccp local FastEthernet0/0  
sccp ccm 192.168.1.10 identifier 1 version 5.0.1  
sccp ip precedence 3  
sccp  
!  
sccp ccm group 1  
  bind interface FastEthernet0/0  
  associate ccm 1 priority 1  
  associate profile 1 register router1-mtp  
!  
dspfarm profile 1 mtp  
  codec pass-through  
  rsvp  
  maximum sessions software 20  
  associate application SCCP  
!  
!  
interface FastEthernet0/1  
  description Site1-WAN  
  ip address 192.168.3.1 255.255.255.0  
  duplex auto  
  speed auto  
  ip rsvp bandwidth 100  
!
```

**IP of CCM.**

**Same name as in CCM MTP**

**Pass-Through codec.**

**100 kbps available bw.**

- ① SCCP and CallManager configuration
- ② SCCP CCM Group
- ③ MTP Profile with RSVP
- ④ Interface configuration with RSVP

# Summary

- **Configure Regions to control codec selection.**
- **Configure Locations to control bandwidth.**
- **RSVP is a hop by hop signaling protocol which reserves bandwidth on a path.**
- **Add an IOS MTP, MRGs and MRGLs to integrate RSVP agents in Cisco CallManager.**
- **Configure IOS gateway to register with Cisco CallManager as IOS MTPs.**

# CISCO SYSTEMS





## **Configuring CallManager for Multi Site Deployments**

# **Configuring Survivable Remote Site Telephony (SRST)**

# Objectives

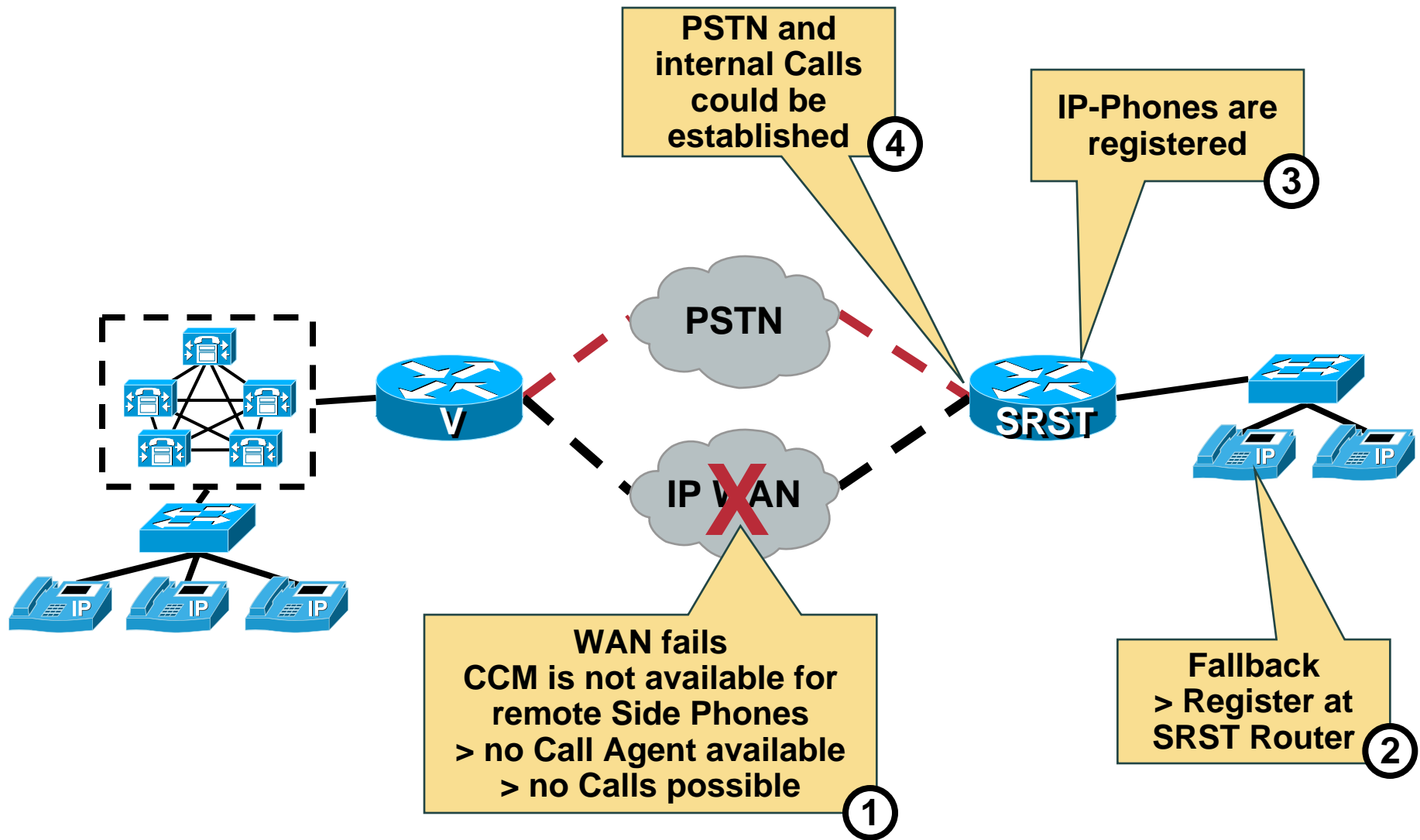
- **Overview of SRST Gateways**
- **SRST Features and Capabilities**
- **SRST Phone Registration Process**
- **Configuring SRST for SCCP Support**
- **Configuring Additional SRST Features for SCCP**
- **Configuring SRST for SIP Support**
- **Configure Cisco Unified CallManager for SRST**
- **Verifying SRST Operation**

# Overview of SRST Gateways

**SRST offers basic phone functionality (Placing and Receiving calls) for remote sites to guard against these possibilities:**

- **Centralized CallManager goes down**
  - WAN failure
  - Server breakdown
- **Different features set for SCCP and SIP SRST**
  - SIP SRST is very basic

# SRST Functionality





# SRST Benefits

- **Business resiliency through redundant, localized call processing**
- **Intelligent and automatic failover configuration—no manual IT or telecom intervention required**
- **Cost-effective operations through a converged voice and data network**
- **Centralized IP Communication configuration and management**
- **Investment protection and ease of migration**

# SRST Gateway Capacity

## SRST gateway capacity depends on:

- Deployed platform
  - CPU
  - Memory
- IOS Version

Visit the following link for more information:

<http://www.cisco.com/univercd/cc/td/doc/product/voice/srst/index.htm>

# Supported Phones

**Most Cisco IP phones are supported when using SRST:**

- **Ensure that all deployed phones in a SRST site are actually supported.**

**Visit the following link for more information:**

**<http://www.cisco.com/univercd/cc/td/doc/product/voice/srst/index.htm>**

# SRST Features

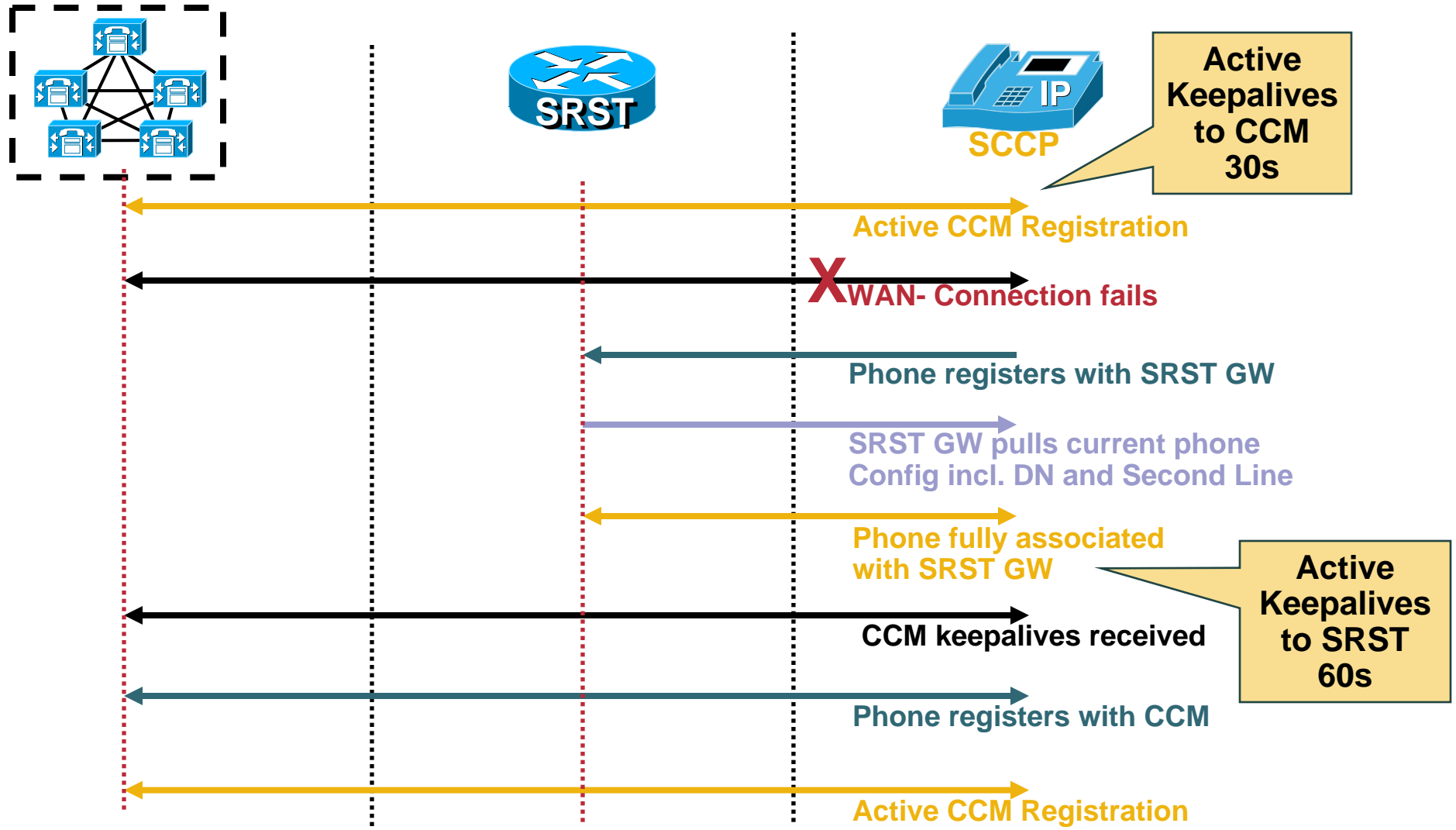
**The following is a partial list of SRST features:**

- Basic phone calls
- Multiple lines per IP phone
- Consultative call transfer, call hold, pickup and hunt groups
- Caller ID information and ANI support
- Additional language options
- Customized system message
- Multicast MoH from flash and Live-feed
- Secure SRST
- Video

**Visit the following link for more information:**

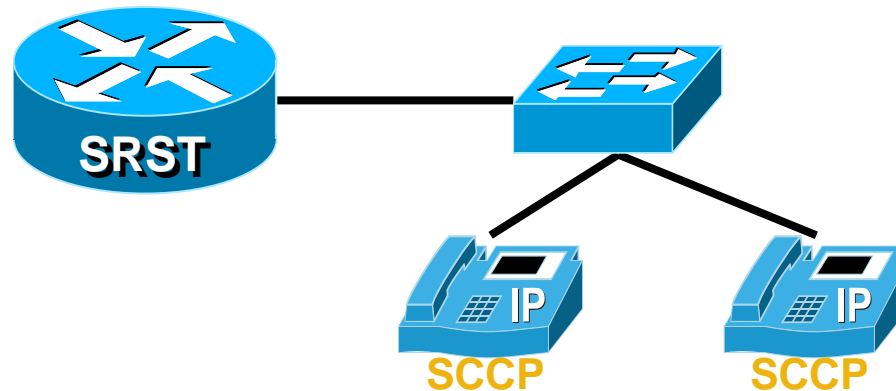
**<http://www.cisco.com/univercd/cc/td/doc/product/voice/srst/index.htm>**

# SRST Phone Registration Process



# Configuring SRST for SCCP Support

1. Enter SRST Configuration mode
2. Define IP Address and Port SRST Service binds to
3. Define Maximum Number of DN to support
4. Define Maximum Number of IP-Phones to support



# SRST Commands

```
router(config)#
```

```
call-manager-fallback
```

- **Enters call-manager-fallback configuration mode.**

```
router(config-cm-fallback)#
```

```
ip source-address ip-address [port port][any-match |  
strict-match]
```

- **Enables the router to receive messages from the Cisco IP phones through the specified IP addresses and provides for strict IP address verification.**
  - **The default port number is 2000**

# SRST Commands (cont.)

```
router(config-cm-fallback)#
```

```
max-dn max-directory-numbers [dual-line] [preference  
preference-order]
```

- **Sets the maximum number of directory numbers (DNs) or virtual voice ports that can be supported by the router and activates the dual-line mode**

```
router(config-cm-fallback)#
```

```
max-ephones max-phones
```

- **Configures the maximum number of Cisco IP phones that can be supported by the router.**



# Sample Router Configuration

## Sample configuration:

```
Router2>ena
Password:
Router2#conf t
Enter configuration commands, one per line.  End with CNTL/Z.
Router2(config)#call-manager-fallback
Router2(config-cm-fallback)# ip source-address 192.168.2.254 port 2000
Router2(config-cm-fallback)# max-ephones 10
Router2(config-cm-fallback)# max-dn 10
Router2(config-cm-fallback)#
```

# Configuring Additional SRST Features for SCCP

```
Router(config-cm-fallback)#
```

```
user-locale country-code
```

- **Selects a language by country for displays on the Cisco IP Phone 7940 and Cisco IP Phone 7960.**

```
Router(config-cm-fallback)#
```

```
max-conferences max-conference-numbers
```

- **Sets the maximum number of simultaneous three-party conferences supported by the router.**

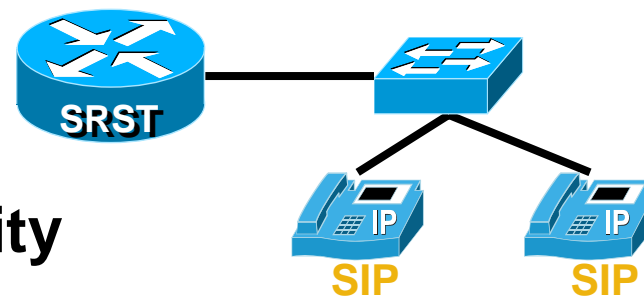
# Sample Router Configuration

## Sample configuration:

```
Password:  
Router2>ena  
Password:  
Router2#conf t  
Enter configuration commands, one per line. End with CNTL/Z.  
Router2(config)#call-manager-fallback  
Router2(config-cm-fallback)# max-conferences 4  
Router2(config-cm-fallback)# user-locale DE
```

# Configuring SRST for SIP Support

1. Enter voice service configuration mode
2. Allow SIP to SIP calls
3. Enter SIP configuration mode
4. Enable SIP registrar functionality
5. Configure a voice register pool
6. Define Phones in register pool
7. Enter voice register global configuration mode
8. Define maximum number of register pools supported



# Configuring SIP SRST commands

```
router(config)#
```

```
voice service voip
```

- Enters voice service configuration mode

```
router(config-voi-srv)#
```

```
allow-connections sip to sip
```

- Enables VoIP-to-VoIP call connections.

```
router(config-voi-srv)#
```

```
sip
```

- Enters SIP configuration mode.

# Configuring SIP SRST commands (cont.)

```
router(conf-serv-sip)#
```

```
registrar server [expires [max sec] [min sec]]
```

- **Enables SIP registrar functionality.**

```
router(config)#
```

```
voice register pool tag
```

- **Enters voice register pool configuration mode for SIP phones**

```
router(config-register-pool)#
```

```
id {network address mask mask | ip address mask mask | mac  
address}
```

- **Explicitly identifies a locally available individual or set of SIP IP phones**

# Configuring SIP SRST commands (cont.)

```
router(config)#
```

```
voice register global
```

- Enters voice register global configuration mode to set global parameters for all supported Cisco SIP IP phones in a Cisco SIP SRST environment.

```
router(config-register-global)#
```

```
max-pool max-voice-register-pools
```

- Sets the maximum number of SIP voice register pools that are supported in a Cisco SIP SRST environment.

# Sample Router Configuration

## Sample configuration:

```
!  
voice service voip  
  allow-connections sip to sip  
  sip  
    registrar server  
!  
voice register global  
  max-pool 10  
!  
voice register pool 1  
  id network 192.168.0.0 mask 255.255.0.0  
!
```



# Configure Cisco Unified CallManager for SCCP SRST

**IP-Phones have to be provided with a SRST Reference to use it as a fallback**

- 1. Add a SRST Reference**
- 2. Assign the SRST Reference to IP-Phones through the Device- Pool**
- 3. Reset Devices to pull the new configuration**

# SRST Reference in Cisco Unified CallManager

The screenshot shows the SRST Reference Configuration page in a Mozilla Firefox browser. The page title is "SRST Reference Configuration - Mozilla Firefox" and the URL is "https://192.168.1.10/ccmadmin/srstSave.do". The page is logged in as "CCMAdministrator".

Four callout boxes provide instructions:

- 1 Use meaningful name**: Points to the "Name\*" field, which contains "CHI SRST".
- 2 Port has to be the same as in the SRST Gateway**: Points to the "Port\*" field, which contains "2000".
- 3 Specify the IP-Address of the SRST Interface**: Points to the "IP Address\*" field, which contains "192.168.2.254".
- 4 Configure IP Address and Port for SIP SRST Support**: Points to the "SIP Network/IP Address" and "SIP Port\*" fields, which contain "192.168.2.254" and "5060" respectively.

The "SRST Reference Information" table is as follows:

SRST Reference	CHI SRST (used by 6 devices)
Name*	CHI SRST
Port*	2000
IP Address*	192.168.2.254
SIP Network/IP Address	192.168.2.254
SIP Port*	5060
SRST Certificate Provider Port*	2445

Additional fields include "SRST Certificate Provider Port\*" (2445) and a checkbox for "Is SRST Secure?". Buttons for "Save", "Delete", "Copy", "Reset", and "Add New" are visible at the bottom. A status message "Update successful" is shown at the top left of the configuration area.

# SRST Reference Association in Device Pools

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The page title is "Device Pool Configuration". The browser address bar shows the URL: https://192.168.1.10/ccmadmin/devicePoolEdit.do?key=b0794fe0-e4d9-ea09-db20-f6c6ec300428. The user is logged in as CCMAdministrator.

Three callout boxes provide instructions:

- Callout 1:** "Use the Device- Pool for the Branch Site" points to the "Device Pool Name" field, which contains "Chicago".
- Callout 2:** "Select the appropriate SRST Reference" points to the "SRST Reference" dropdown menu, which is set to "CHI SRST".
- Callout 3:** "Reset the Device Pool to refresh the configuration on the Devices" points to the "Refresh" button (a circular arrow icon) in the top left toolbar.

The "Device Pool Settings" table is as follows:

Field	Value
Device Pool Name*	Chicago
Cisco CallManager Group*	Default
Date/Time Group*	CMLocal
Region*	CHI
Softkey Template*	Standard User
SRST Reference*	CHI SRST
Calling Search Space for Auto-registration	< None >
Media Resource Group List	< None >
Network Hold MOH Audio Source	< None >
User Hold MOH Audio Source	< None >

# Verifying SRST Operation

router#

```
show running-config
```

- **command to verify the configuration**

router#

```
show call-manager-fallback all
```

- **command to verify that the Cisco SRST feature is enabled.**

router#

```
show ephone
```

- **command to display the Cisco IP phones that have registered to the Cisco SRST router**

# Verifying SRST Operation (cont.)

router#

```
debug ephone register
```

- **command to display the Cisco IP phones that have registered to the Cisco SRST router**

## Verifying on IP- Phones

- **Use network Settings display on the Cisco IP phones to verify that the SRST Device on the phones matches the IP address of the Cisco SRST router**
- **Cisco IP phones display a message informing you that they are operating in Cisco Unified CallManager fallback mode**

# Verifying SRST Operation (cont.)

## Sample configuration:

```
Router2#show running-config
Building configuration...
!
!
call-manager-fallback
  max-conferences 4 gain -6
  user-locale DE
  ip source-address 192.168.2.254 port 2000
  max-ephones 10
  max-dn 10
!
!
```

# Summary

- **SRST is a way in Cisco IP Communication Environments to ensure operation of the remote branch site phones even if the Cisco Unified CallManager is not available**
- **Basic Telephony Features could be assured**
- **In Fallback mode the Gateway pulls the current Configuration of the IP Phones**
- **For Basic Operation only a IP Address and port where service binds to have to be defined.**
- **Additionally the maximum of SRST Phones and DN's have to be configured cause per default they are zero.**

## Summary (cont.)

- **SRST Features are mainly derived from the CCME Feature set.**
- **For operating as a SIP SRST device the registrar service on the router has to be enabled.**
- **In Cisco Unified CallManager a SRST Reference is added and then provided to the IP phones through the Device Pool.**
- **IP Access Lists can help verify SRST operation, but in Lab Environment just shutting down the CallManager Service may be the easier way.**



# CISCO SYSTEMS





## **Cisco Unified CallManager 5.0 Features and Applications**

# **Introducing Cisco Unified CallManager 5.0 Applications**

# Objectives

- **Application Overview**
- **Cisco Unified CallManager 5.0 Out-of-the-Box Applications**
- **IP Manager Assistant Overview**
- **Extension Mobility Overview**
- **Attendant Console Overview**
- **Presence Overview**
- **Cisco Applications**
- **3rd Party Applications**

# Application Overview

**Cisco Unified CallManager is a full featured, application rich Unified Communication System:**

- **Comes with applications that cover the need of today's communications**
- **Seamlessly integrated with Cisco Unified CallManager**
- **Integrated in system testing and development**
- **There is no additional licensing needed**
- **All applications that are built in could be used at no additional cost**

# Cisco Unified CallManager 5.0

## Out-of-the-Box Applications

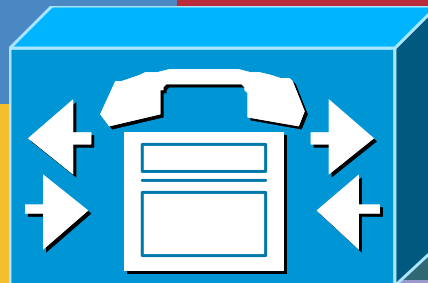
### IPMA

Manager Assistant Application  
to enhance productivity and  
cover needs of executive staff

### Extension

### Mobility

IP Phone Service  
to cover mobility needs



### Web Dialer

Click to Dial Application  
for end-users

### Attendant Console

PC Application  
to cover the needs of the  
receptionist or switchboard

### 3rd Party Applications

API  
interfaces  
to support 3rd Party  
Application Developments

# IP Manager Assistant Overview

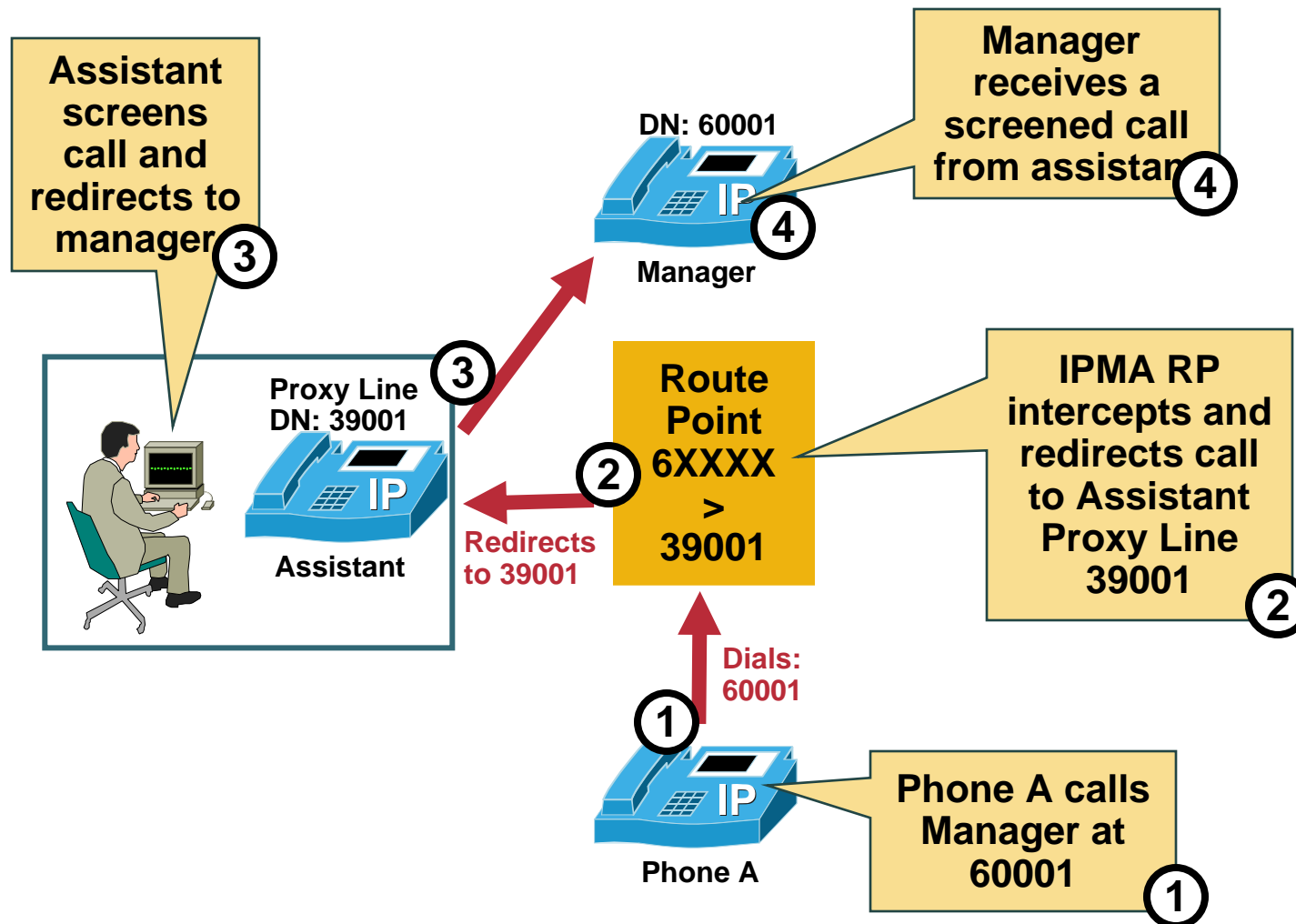
**Enables managers and assistants to work together more effectively**

- **Enhanced phone capabilities for the manager and assistant.**
- **Available desktop interfaces that are primarily used by the assistant.**

**Cisco IPMA supports two modes of operation:**

- **proxy line mode**
- **shared line mode**

# IPMA Overview: Proxy Line Mode







# IPMA Overview

## Features for Managers Phones:

- **Use these Softkeys**
  - **ImmDiv** - Immediately diverts the Call
  - **DivAll** - Diverts all Calls
  - **DND** - Mute ring tone immediately

## Proxy line mode features

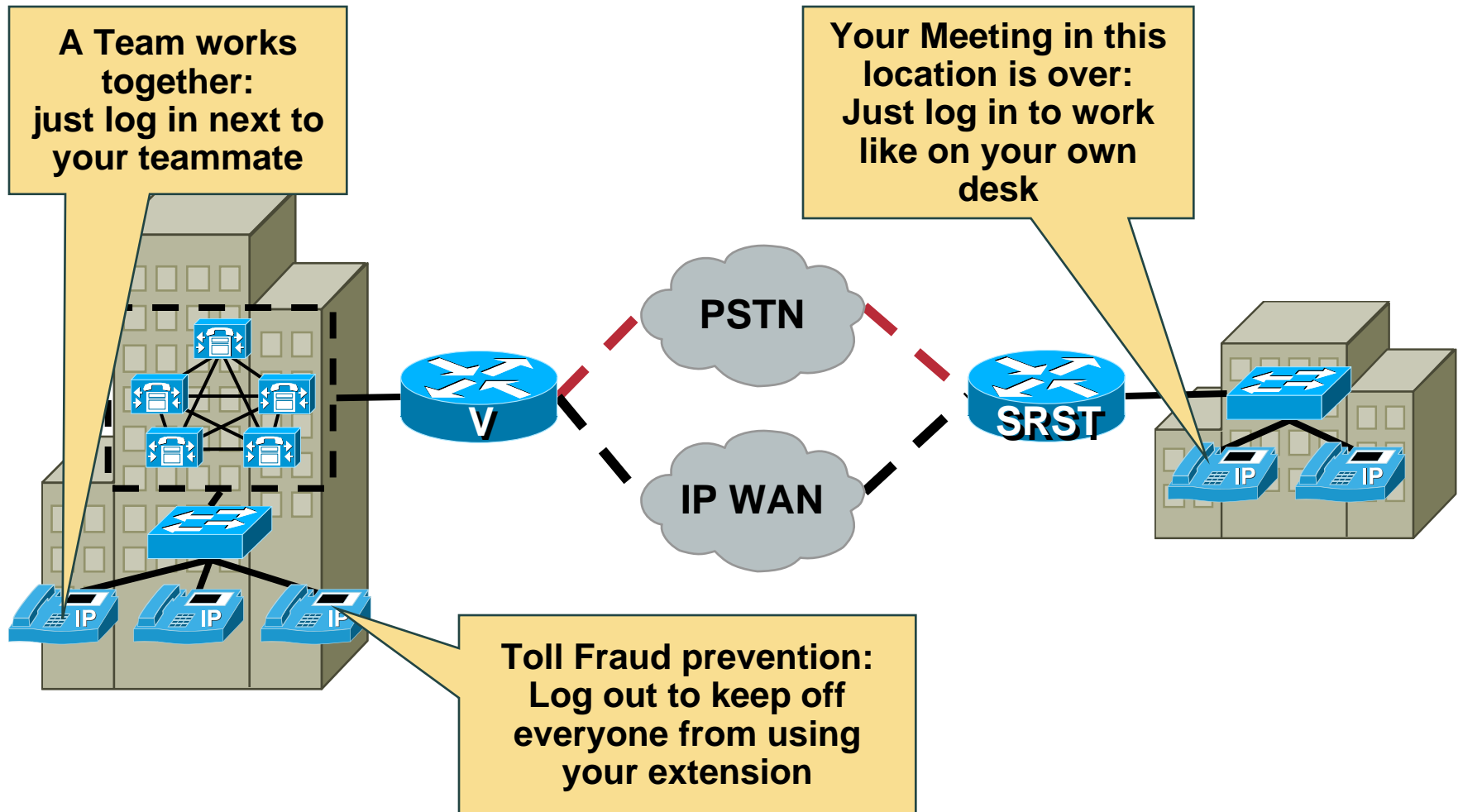
- **Call routing services are unique to proxy line mode**
  - **Call filtering, call intercept, assistant watch, and assistant selection**
  - **SetWtch** - Monitor Assistant Line

# Extension Mobility Overview

## **Extension Mobility is an IP phone Service**

- **It is available through the service button**
- **Use your userID and PIN for authentication**
- **Users can get their usual extension speed-dials and services by logging in to an IP phone**
- **Users can prevent toll-fraud by logging out their phones**
- **You are able to work and to be available everywhere in your companies IP network under your own DN**

# Extension Mobility Usage Examples



# Extension Mobility Overview

## SCCP versus SIP Phone Support

- Support for EM on SCCP phones  
7970/71, 7960/7961, 7940/41, 7920, 7912/11/05
- Support for EM on SIP phones is limited to enhanced phones  
7970/71, 7961/41, 7911

## New Features with CCM 5.0

- Ability to clear the call logs (missed, received, dialed) during manual EM login and Logout
- Can be configured using a cluster wide service parameter

# Extension Mobility Overview

## Restrictions/ Caveats:

- EM is supported only within a single cluster, there is no support for multiple/distributed call processing deployments.
- EM users should not move between locations or sites within a cluster when using AAR or the VoPSTN deployment model.
- EM functionality still remains dependent on the Publisher (if it is down users can not login or logout of a phone).

**These restrictions have not changed  
They exist with CM 4.x too**

# Attendant Console Overview






**Tool for enterprise attendants and receptionists to answer and greet callers, then efficiently dispatch calls**

- **PC based Application**
- **Calls can be forwarded or transferred by either an application GUI or an attendants phone**
- **PC application is capable of monitoring line states**
- **Hunt Groups can distribute calls to groups of attendants**
- **Distribution to Hunt Group Members is either circular or broadcast**
- **Queuing can be enabled at pilot points**

# Attendant Console Overview Client GUI Summary

## Drag-and-drop hold, transfer

### Line states

- Idle line 
- Inbound call 
- Hold 
- Active 
- Unknown 

### Directory Pane

- Search by any field in the directory
- Sortable columns
- Call Forward status icons

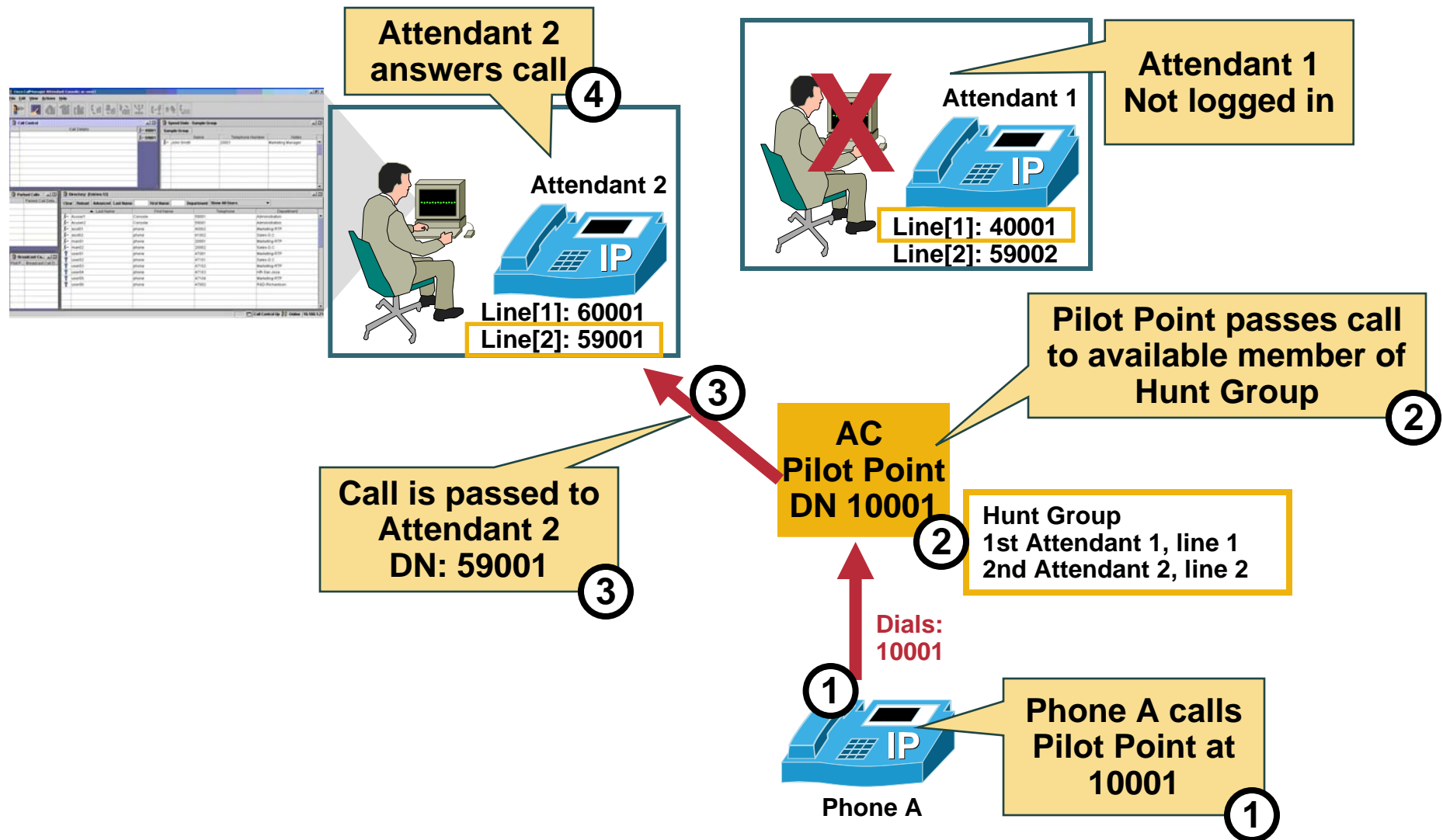


Name	Telep...	Notes
Sales	23213	
Warehouse	24758	
HR	29854	
Payroll	28435	

Last Name	First Name	Telephone	Department
Carter	Jeff	1011	Executive
Chambers	Debby	1010 →	Executive
Cheeniyl	Greg	1001 ☒	Engineering
lowery	Bryan	1003	Marketing
Prabhakar	Cindy	1012	Engineering
Tikku	Julia	1000	Engineering
Vankayala	Baja	1002	Engineering

# Attendant Console Overview: Hunt Groups





# Attendant Console Overview

## Restrictions:




- **AC is only supported for these SCCP phones (7970/71, 7960/7961, 7940/41, 7912/11/05)**
- **AC is not supported on SIP phones (existing or enhanced)**
- **Hunt group members or AC Pilot Points cannot share lines**
- **Hunt Group members cannot use overlapping extensions**

# Presence Overview - Introduction

- **Presence – Status of a device that is used for communication**  
**Examples:**
  - Phone on-hook,
  - Busy
  - Unknown
- **Cisco Unified CallManager 5.0 supports BLF – Busy Lamp Field:**
  - Visible indication of on-hook and busy state.
  - A phone icon or a light that turns on/off

# Busy Lamp Field (BLF) Feature

- BLF is configured in the phone button template
- Works like SpeedDial button
- BLF states are indicated by icons and/or LEDs

State	Icon	LED
Idle		Off
Busy		Steady Red
Unknown		Off

# BLF Feature Phone Support for Call Manager 5.0

Cisco IP Phones	BLF Status for Call Lists Support	BLF Status for Speed Dial Support
7902, 7905, 7910, 7920, 7935, 7936, 7989, ATA, VG 248	No	No
7914, 7940, 7960 SCCP	No	Yes
7970 SIP and SCCP	Yes	Yes
Enhanced Cisco IP Phones – 7941, 7961, 7971 SIP and SCCP (not released yet)	Yes	Yes
IP Communicator	No	No

# Additional Cisco Applications

## **Other Cisco Unified CallManager end user communication enhancing applications:**

- **Web Dialer:**  
Click to Dial application accessible through a browser
- **Personal Directory Applications:**
  - **Personal Address Book (PAB)**
  - **Personal Fast Dials (FastDials)**
  - **Administered through the user webpage by the user to service the need for personal directories.**

# 3rd Party Applications

## 3rd Party Application Development is driven by API's available from Cisco

- **Protocols supported with CCM:**
  - **SOAP/ AXL/ XML /TAPI /JTAPI**
- **Developer Guides are available**
- **Traces could be used for troubleshooting**

# Summary

- **Cisco Unified CallManager is a full featured application rich Unified Communication System that comes with all application available at no additional cost.**
- **There are many applications already built in plus standard interfaces to support 3rd party applications**
- **IP Manager Assistant Application could be run in proxy and in shared line mode**
- **Extension Mobility enables users to have their extension travel with them.**
- **Attendant Console enables switchboard staff to quickly forward and transfer calls.**
- **Presence is the upcoming next step in today's communication and Cisco Unified CallManager takes it.**
- **API's with the support of standard protocols allow 3rd party application development**

# CISCO SYSTEMS







# Cisco Unified CallManager 5.0 Features and Applications

## Configuring Extension Mobility

# Objectives

- **Extension Mobility Configuration Steps**
- **Extension Mobility Service Configuration**
- **Extension Mobility Configuration on IP Phone**
- **Create Profiles**
- **Configure User for Extension Mobility**

# Extension Mobility Configuration Steps

## Overview of Configuration Steps

- 1. Service Activation**
- 2. Create EM IP phone service**
- 3. Create user device profile**
- 4. Associate device profile to user**
- 5. Enable EM and configure default device profile on the target device**
- 6. Subscribe to EM IP phone service on the target device and the user device profile**

# Extension Mobility Service Configuration

**To enable extension mobility on Cisco Unified CallManager 5.0 the following tasks need to be performed:**

- **Enable extension mobility service.**
- **Configure extension mobility service parameters.**
- **Add extension mobility device service.**

# Extension Mobility Service Configuration: Enable Service

The screenshot shows the Cisco CallManager Serviceability web interface in a Mozilla Firefox browser window. The page title is "Cisco CallManager Serviceability - Service Activation". The browser address bar shows the URL: `https://192.168.1.10/ccmservice/serviceactivation.jsp?htxtNodeID=8fb74274-671f-491a-93e3-afb5ee74ed73&htxtSubmit=Update&publisherNode=t`. The page header includes "Navigation" and "Cisco CallManager Serviceability". Below the header, there are menu items: "Alarm", "Trace", "Tools", "Snmp", and "Help". The main content area is titled "Service Activation" and includes a "Related Links" section with a dropdown menu set to "Control Center - Feature Services" and a "Go" button. Below this, there are icons for "Status", "Select Server", and "CM Services". The "Status" section shows "Update Completed". The "Select Server" section has a dropdown menu set to "192.168.1.10". The "CM Services" section contains a table with the following data:

Service Name	Activation Status
<input checked="" type="checkbox"/> Cisco CallManager	Activated
<input checked="" type="checkbox"/> Cisco Tftp	Activated
<input type="checkbox"/> Cisco Messaging Interface	Deactivated
<input checked="" type="checkbox"/> Cisco IP Voice Media Streaming App	Activated
<input checked="" type="checkbox"/> Cisco CTIManager	Activated
<input type="checkbox"/> Cisco CallManager Attendant Console Server	Deactivated
<input checked="" type="checkbox"/> Cisco Extension Mobility	Activated
<input checked="" type="checkbox"/> Cisco Extended Functions	Activated
<input checked="" type="checkbox"/> Cisco CallManager Cisco IP Phone Services	Activated
<input type="checkbox"/> Cisco Dialed Number Analyzer	Deactivated

A red box highlights the "Cisco Extension Mobility" row, and a yellow callout bubble with the text "Activate EM service" points to it. The browser status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Extension Mobility Service Configuration: Service Parameters

**Service Parameter Configuration - Mozilla Firefox**

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/serviceParamEdit.do?server=8fb74274-671f-491a-93e3-afb5ee74ed73&service=8

**Service Parameter Configuration** Related Links: Parameters for All Servers Go

**Select EM service** ①

Status  
Status: Ready

Select Server and Service

Server \* 192.168.1.10 (Active)

Service \* Cisco Extension Mobility (Active)

All parameters apply only to the current server except parameters that are in the Clusterwide group

**Configure EM service parameters** ②

**Cisco Extension Mobility (Active) Parameters on server 192.168.1.10 (Active)**

Parameter Name	Parameter Value	Suggested Value
<b>Clusterwide Parameters (Parameters that apply to all servers)</b>		
<a href="#">Enforce Maximum Login Time *</a>	False	False
<a href="#">Maximum Login Time *</a>	8:00	8:00
<a href="#">Maximum Concurrent Requests *</a>	3	3
<a href="#">Multiple Login Behavior *</a>	Multiple Logins Not Allowed	Multiple Logins Not Allowed
<a href="#">Alphanumeric User ID *</a>	True	True
<a href="#">Remember the Last User Logged In *</a>	False	False
<a href="#">Clear Call Log *</a>	False	False

Save Set to Default Advanced

Done 192.168.1.10

# Extension Mobility Service Configuration: Service Parameters

Field	Settings	Description
Enforce max. logon time	False, True	Enable/disable auto logout after defined time.
Max. logon time	Up to 168:00 hours	Define max. logon time.
Max. concurrent requests	Allowed, Not allowed, Auto Logout	Define if users can/cannot log on to multiple phones, or are logged of from already logged in phone.
Alphanumeric user ID	False, True	Define if alphanumeric user IDs are allowed.
Remember latest logged in	False, True	Define if phones remember last user logged in on EM service logon.
Clear call log	False, True	Define if missed, placed, and received call directories are stored if logging on/off.

# Extension Mobility Service Configuration: IP Phone Service

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays the URL: `https://192.168.1.10/ccmadmin/phoneServicesFindList.do?%3C%=reqParams%3E&recCnt=0&colCnt=3`. The page title is "Find and List IP Phone Services". The navigation menu includes "Cisco CallManager Administration" and "Go". The user is logged in as "CCMAdministrator". The main content area shows "Find and List IP Phone Services" with a plus sign icon. Below this, there is a "Status" section indicating "0 records found". The "Search Options" section includes a search form with a dropdown menu set to "IP Phone Service", a "begins with" dropdown, a text input field, a "Find" button, and a checkbox for "Search Within Results". The "Search Results" section displays the message "No active query. Please enter your search criteria using the options above." and includes an "Add New" button (highlighted with a red box) and a "Rows per Page" dropdown set to "50". A yellow callout bubble with the text "Add a new IP phone service" and a circled "1" points to the "Add New" button.



# Extension Mobility Service Configuration: IP Phone Service

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The page title is "IP Phone Services Configuration". The browser address bar shows the URL "https://192.168.1.10/ccmadmin/phoneServicesEdit.do". The navigation menu includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The main content area is titled "IP Phone Services Configuration" and contains a "Save settings" button (callout 4) and a "Save" button. The "Service Information" section is highlighted with a red box and contains the following fields:

Service Name *	ASCII Service Name *
Logon/Logoff	Logon/Logoff
Service Description	
Extension Mobility Logon/Logoff	
Service URL *	
http://192.168.1.10:8080/emapp/EMAppServlet?device=#DEVICENAME#	

Callout 2 points to the "Service Name" and "Service Description" fields. Callout 3 points to the "Service URL" field. Callout 4 points to the "Save settings" button. A legend indicates that an asterisk (\*) indicates a required item.

# Extension Mobility Configuration on IP Phone

**To enable extension mobility on Cisco IP phones the following tasks need to be performed:**

- **Subscribe extension mobility phone service.**
- **Enable extension mobility on the device.**

# Extension Mobility Configuration on IP Phone

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows `https://192.168.1.10/ccmadmin/phoneSave.do`. The page title is "Phone Configuration - Mozilla Firefox". The navigation menu includes "Cisco CallManager Administration" and "Go". The user is logged in as "CCMAdministrator". The main menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", "Help", and "Log Off". The "Phone Configuration" section is active, and the "Subscribe/Unsubscribe Services" dropdown menu is highlighted with a red box. A yellow callout bubble with the text "Subscribe IP phone service" and a circled "1" points to the "Subscribe" option in the dropdown. The "Status" section shows "Update successful". The "Association Information" section lists 11 items, including "Line [1] - 3001 (no partition)", "Line [2] - Add a new DN", and "Add a new SD" (items 3-8). The "Phone Type" section shows "Product Type: Cisco IP Communicator" and "Device Protocol: SCCP". The "Device Information" section shows "Registration: Registered with Cisco CallManager 192.168.1.10", "IP Address: 10.128.128.138", "Device Name\*: SEP000BDBDF61DE", "Description: SJC Phone 1", "Device Pool\*: Default", "Phone Button Template\*: Default IP Communicator Template", "Softkey Template: < None >", "Common Phone Profile\*: Standard Common Phone Profile", "Calling Search Space: < None >", and "AAR Calling Search Space: < None >".

# Extension Mobility Configuration on IP Phone

The screenshot displays the Cisco CallManager administration interface in a Mozilla Firefox browser window. The page title is "Subscribed Cisco IP Phone Services for SEP000BDBDF61DE". The interface includes a left-hand navigation menu with "Phone Configuration" selected. The main content area shows the configuration for a specific phone service. A red box highlights the "Logon/Logoff" service selected in the "Select a Service" dropdown menu. A yellow callout bubble with a circled "2" points to this dropdown, containing the text "Select EM service". Below this, a "Next" button is highlighted with a red box, and a yellow callout bubble with a circled "3" points to it, containing the text "Click next.". The status bar at the bottom of the browser window shows "Done" and the URL "https://192.168.1.10".

# Extension Mobility Configuration on IP Phone

The screenshot displays the Cisco CallManager Phone Configuration web interface. The browser window title is "Phone Configuration - Mozilla Firefox" and the address bar shows "https://192.168.1.10/ccmadmin/phoneSave.do". The page content includes a sidebar with "Cisco CallMan" and "Phone Configuration" sections. The main content area is titled "Subscribed Cisco IP Phone Services for SEP000BDBDF61DE" and contains the following sections:

- Status:** Status: Ready
- Service Information:** Service Name: Logon/Logoff, Service Name \*: Logon/Logoff, ASCII Service Name \*: Logon/Logoff
- Subscribed Services:** A table with a "Subscribe" button highlighted by a red rectangle and a yellow callout box labeled "4".

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# Extension Mobility Configuration on IP Phone

Phone Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/phoneEdit.do?key=2b5be699-488f-4e43-b518-52f0e24ac0f6

Services

Authentication Server

Proxy Server

Idle

Idle Timer (seconds)

**Extension Information**

Enable Extension Mobility

Log Out Profile [- Use Current Device Settings -]

Login in User ID < None >

Log in Time < None >

Log out Time < None >

**MLPP Information**

MLPP Domain < None >

MLPP Indication\* Default

**Secure Shell Information**

Secure Shell User

Secure Shell Password

**Product Specific Configuration**

Auto Line Select \* Disabled

IP Address

Autodetection URL

LD&P Server Information

Done 192.168.1.10

**Enable EM on the phone** 5

# Create Profiles

**Creating device profiles used by extension mobility is similar to adding new phones:**

- **Add a new profile.**
- **Select device type.**
- **Configure device.**
- **Configure line.**
- **Add extension mobility service.**

# Create Profiles

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10/ccmadmin/deviceProfileFindList.do`. The page title is "Find and List Device Profiles". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The "Find and List Device Profiles" section contains a search form with the following fields:

- Find: User
- Device Profile where: Profile Name
- begins with: [empty]
- Find button
- Search Within Results

The search results area displays the message: "No active query. Please enter your search criteria using the options above." Below this message is an "Add New" button, which is highlighted with a red box. A yellow callout bubble with the text "Add a new device profile" and a circled "1" points to the "Add New" button.



# Create Profiles

The screenshot shows the Cisco CallManager Administration interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10/ccmadmin/deviceProfileEdit.do`. The page title is "Device Profile Configuration" and it is logged in as "CCMAdministrator".

Annotations on the page include:

- A yellow callout bubble with the text "Click next" and a circled number "3" pointing to the "Next" button.
- A yellow callout bubble with the text "Select phone type" and a circled number "2" pointing to the "Device Profile Type" dropdown menu.

The "Device Profile Type" dropdown menu is open, showing a list of phone models:

- Cisco IP Communicator
- Cisco 7905
- Cisco 7911
- Cisco 7912
- Cisco 7920
- Cisco 7940
- Cisco 7941
- Cisco 7941G-GE
- Cisco 7960
- Cisco 7961
- Cisco 7961G-GE
- Cisco 7970
- Cisco 7971
- Cisco 7985
- Cisco IP Communicator

The "Next" button is highlighted with a red box. A status message at the bottom of the page reads "Done". The system status bar at the bottom right shows the IP address "192.168.1.10".

# Create Profiles

The screenshot shows the 'Device Profile Configuration' page in a Mozilla Firefox browser. The page title is 'Device Profile Configuration' and the URL is 'https://192.168.1.10/ccmadmin/deviceProfileEdit.do?clone=1&key=997642dd-765c-b045-7a93-935e6c2d2d59'. The page contains several sections:

- Status:** Status: Ready
- User Device Profile Information:**
  - Product Type: Cisco IP Communicator
  - Device Protocol: SCCP
  - Device Profile Name\*: legordon
  - Description: EM profile Ed Gordon
  - User Hold Audio Source: < None >
  - User Locale: < None >
  - Phone Button Template\*: Default IP Communicator Template
  - Softkey Template: < None >
  - Ignore Presentation Indicators (internal calls only)
- Multilevel Precedence and Preemption (MLPP) Information:**
  - MLP: [dropdown]
  - MLP: [dropdown]
- Logged (Default) Profile Information:**
  - Login User Id: < None >

At the bottom left, there is a 'Save' button. At the bottom right, the status bar shows 'Done' and the IP address '192.168.1.10'.

Numbered callouts are present:

- 4:** Enter name and description for the profile (points to the 'Device Profile Name' and 'Description' fields).
- 5:** Select phone button template (points to the 'Phone Button Template' dropdown).
- 6:** Create profile (points to the 'Save' button).

# Create Profiles

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10/ccmadmin/deviceProfileSave.do`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The main content area is titled "Device Profile Configuration" and includes a "Related Links" section with a "Back To Find/List" button. Below this, there are icons for save, delete, and add.

The "Status" section shows "Add successful".

The "Association Info" section contains a list of items:

- 1 [Line \[1\] - Add a new DN](#) (highlighted with a red box)
- 2 [Line \[2\] - Add a new DN](#)
- 3 [Add a new SD](#)
- 4 [Add a new SD](#)
- 5 [Add a new SD](#)
- 6 [Add a new SD](#)
- 7 [Add a new SD](#)
- 8 [Add a new SD](#)
- 9 [Add a new SD](#)
- 10 [Add a new SURL](#)
- 11 Privacy

The "Information" section contains the following fields:

- Product Type: Cisco IP Communicator
- Device Protocol: SCCP
- Device Profile Name\*: egordon
- Description: EM profile Ed Gordon
- User Hold Audio Source: < None >
- User Locale: < None >
- Phone Button Template\*: Default IP Communicator Template
- Softkey Template: < None >
- Ignore Presentation Indicators (internal calls only)

The "Multilevel Precedence and Preemption (MLPP) Information" section contains:

- MLPP Domain: < None >
- MLPP Indication\*: Off

A yellow callout box with a circled "7" points to the "Add a new DN" link in the Association Info list. The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Create Profiles

The screenshot displays the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL: `https://192.168.1.10/ccmadmin/directoryNumberEdit.do?devicekey=997642dd-765c-b045-7a93-935e6c2d2d59&index=1&routepartitionkey=35d43f3e`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator".

The main content area is titled "Directory Number Configuration" and includes a "Related Links" section with a "Back To Find/List" button. Below this, there is a "Status" section showing "Status: Ready" and a note: "Note: Changes to Line or Directory Number settings require restart."

The "Directory Number Information" section contains the following fields:

- Directory Number\*: 2700
- Route Partition: SJC-Phones
- Description: Line Ed Grodon
- Alerting Name: Ed Gordon
- ASCII Alerting Name: Ed Gordon
- Active

The "Directory Number Settings" section contains the following fields:

- Voice Mail Profile: < None > (Choose <None> to use system default)
- Calling Search Space: < None >
- Presence Group\*: Standard Presence group
- AAR Group: < None >
- User Hold Audio Source: < None >

A yellow callout bubble with the text "Configure line" and a circled number "8" points to the "Route Partition" dropdown menu.

# Create Profiles

The screenshot shows the Cisco CallManager Administration interface in Mozilla Firefox. The browser address bar shows the URL: `https://192.168.1.10/ccadmin/deviceProfileEdit.do?key=997642dd-765c-b045-7a93-935e6c2d2d59`. The page title is "Device Profile Configuration".

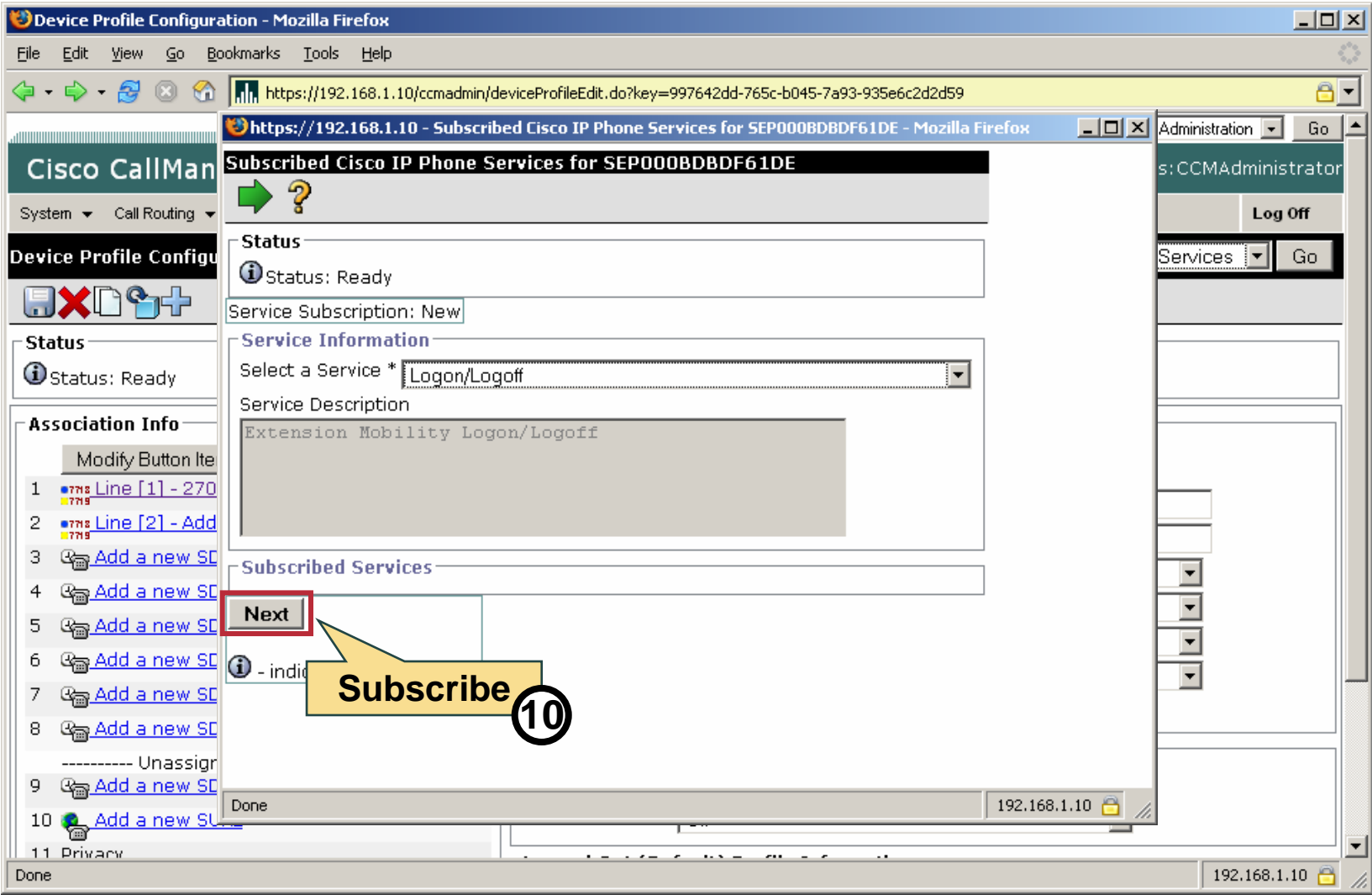
The navigation menu includes: System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, Help, and Log Off. A "Related Links" section contains a dropdown menu for "Subscribe/Unsubscribe Services" and a "Go" button, which is highlighted by a red box and a yellow callout bubble containing the number 9.

The main content area is divided into two sections:

- Association Info:** A list of 11 items for association, including "Line [1] - 2700 in SJC-Phones", "Line [2] - Add a new DN", and several "Add a new SD" and "Add a new SURL" options.
- User Device Profile Information:** Fields for Product Type (Cisco IP Communicator), Device Protocol (SCCP), Device Profile Name (egordon), Description (EM profile Ed Gordon), User Hold Audio Source, User Locale, Phone Button Template (Default IP Communicator Template), and Softkey Template. There is also a checkbox for "Ignore Presentation Indicators (internal calls only)".

At the bottom, the "Multilevel Precedence and Preemption (MLPP) Information" section shows MLPP Domain as "< None >" and MLPP Indication\* as "Off".

# Create Profiles



# User Configuration

**To allow users to log on with the configured device profile, the following step need to be performed:**

- **Add the device profile to the user.**

# User Configuration

Find and List Users - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/userFindList.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

Find and List Users

+ [Grid Icon] [Refresh Icon] [Close Icon]

Status

1 records found

Search Options

Find user where User ID begins with egordon Find Search Within Results

(userid begins with egordon)

Search Results

	User ID	First Name	Last Name	Department
<input type="checkbox"/>	egordon	Ed	Gordon	

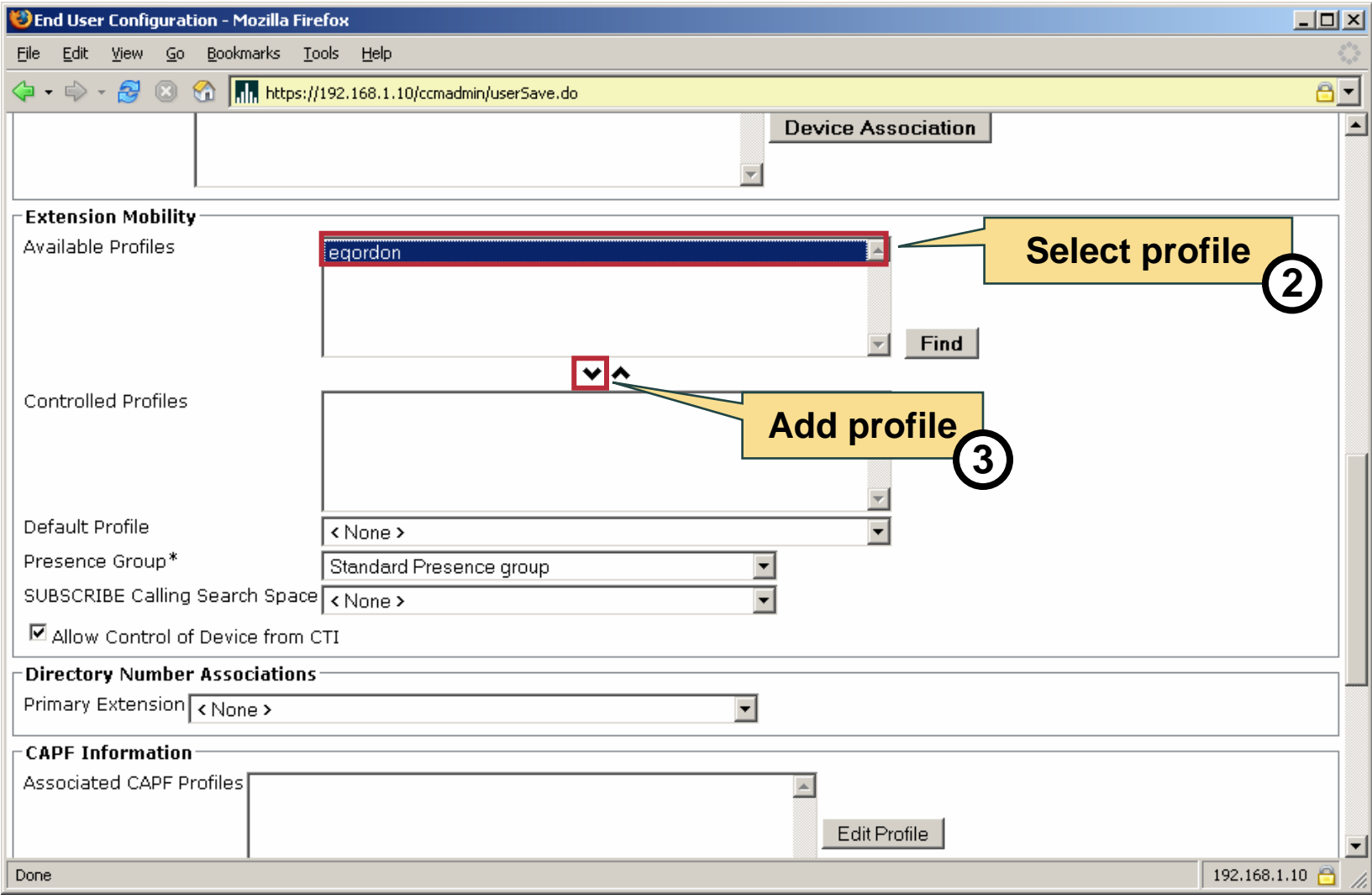
Add New Select All Clear All Delete Selected Rows per Page 50

Select the user ①

Done 192.168.1.10



# User Configuration



# Summary

- **The extension mobility service needs to be enabled on Cisco Unified CallManager 5.0 and configured with custom specific values.**
- **Extension mobility needs to be enabled on every phone that should support that feature.**
- **Each user that should use extension mobility in a Cisco Unified CallManager 5.0 environment needs his own device profile.**
- **The device profile needs to be added to the controlled extension mobility profiles of the user.**

# CISCO SYSTEMS





# Cisco Unified CallManager 5.0 Features and Applications

## Presence and Busy Lamp Field

# Objectives

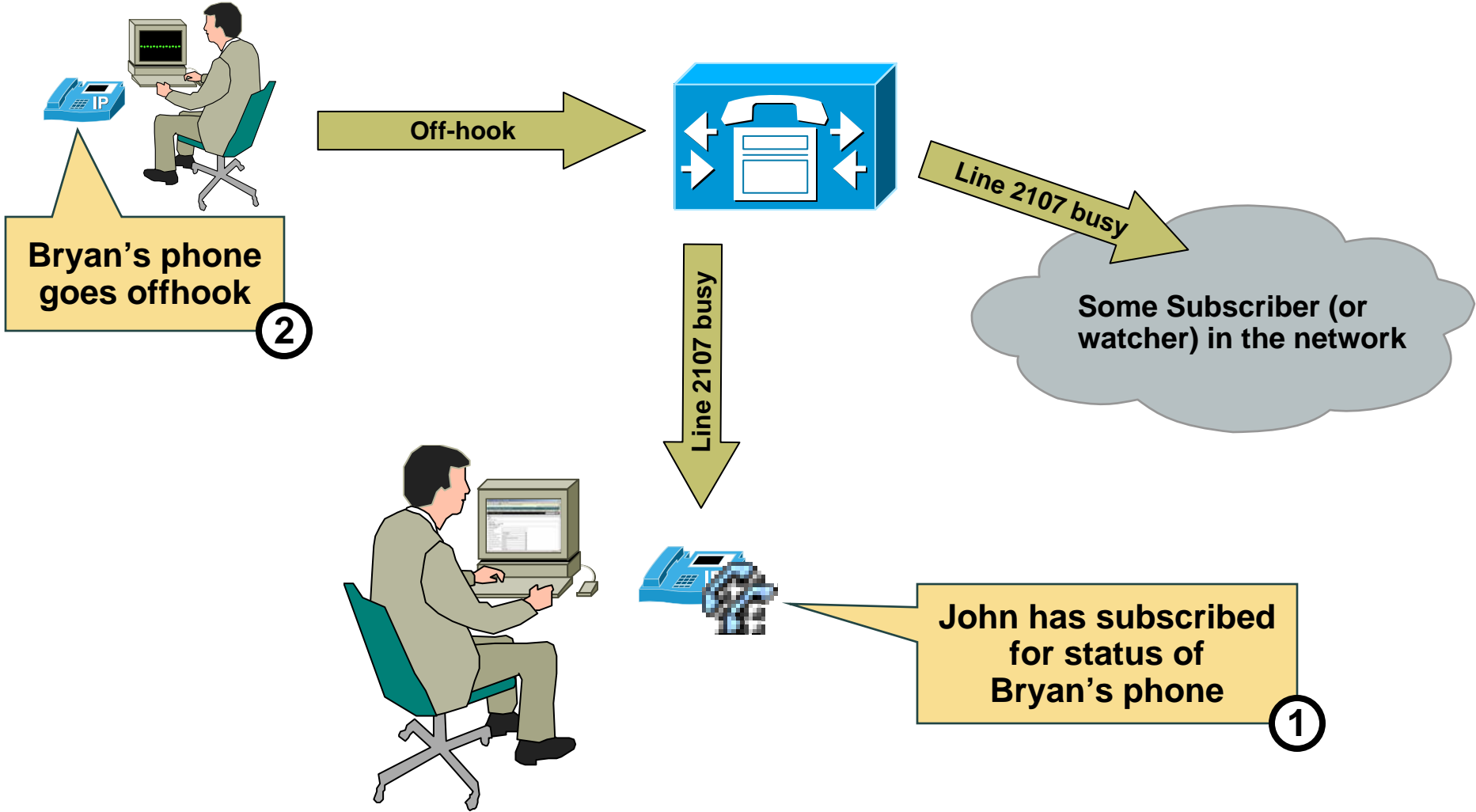
- **Speed Dial BLF and Call List BLF**
- **Configure BLF**
- **Configure Phones for BLF**
- **Configure Presence Authorization**

# Speed Dial BLF and Call List BLF

**Cisco Unified CallManager 5.0 presence implementation follows these guidelines:**

- **Conformant to the SIP standards for SIMPLE - SIP for Instant Messaging and Presence Leveraging Extensions (IETF RFC 3265, 3856, 3863).**
- **Skinny side uses SCCP extensions for similar concepts of subscription and notification.**
- **Allows subscriptions to CM line status and notify when the status changes.**
- **Allows subscriptions for line status between SIP and SCCP phones on Call Manager.**
- **Allows subscriptions from a CM endpoint to an external SIP endpoint (or vice versa) using SIP trunk.**

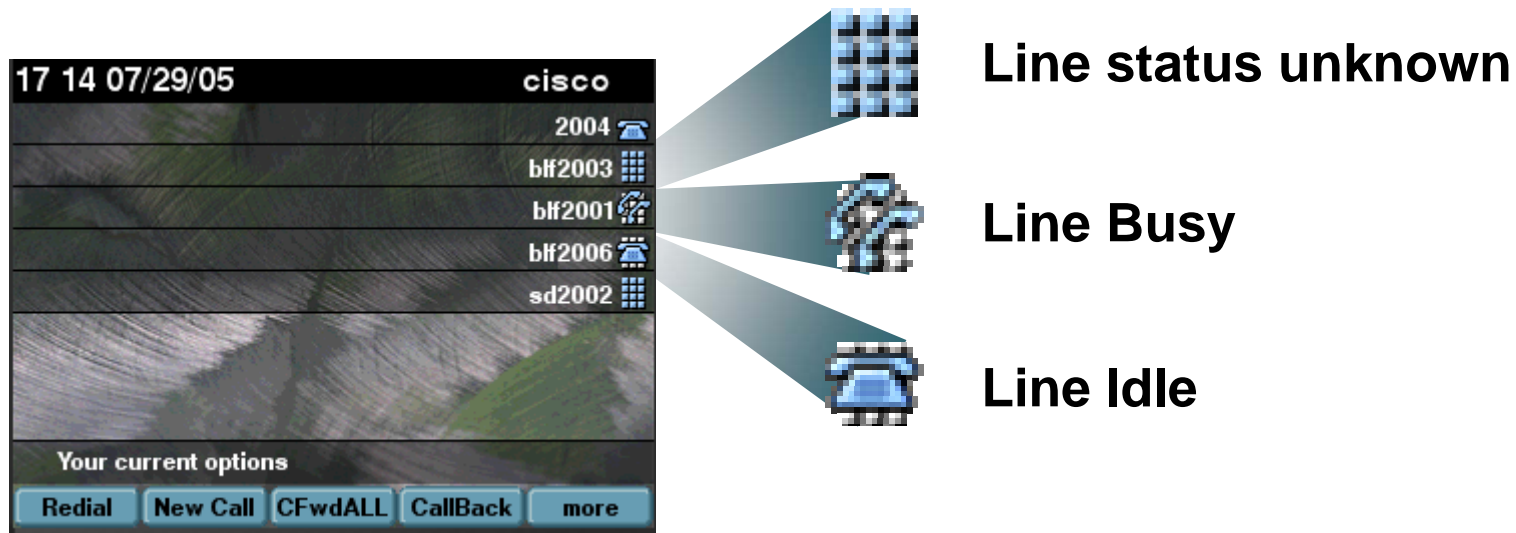
# Speed Dial BLF and Call List BLF



# Speed Dial BLF and Call List BLF

## Speed dial BLF:

- Permanently displaying the status of the speed dial DN or URL.
- Is supported on Cisco IP phones 7940 and greater.





# Speed Dial BLF and Call List BLF

## Call list BLF:

- Displaying the status of the DN or URL on missed, received, placed calls and directory.
- Is supported on Enhanced Cisco IP phones and 7970.



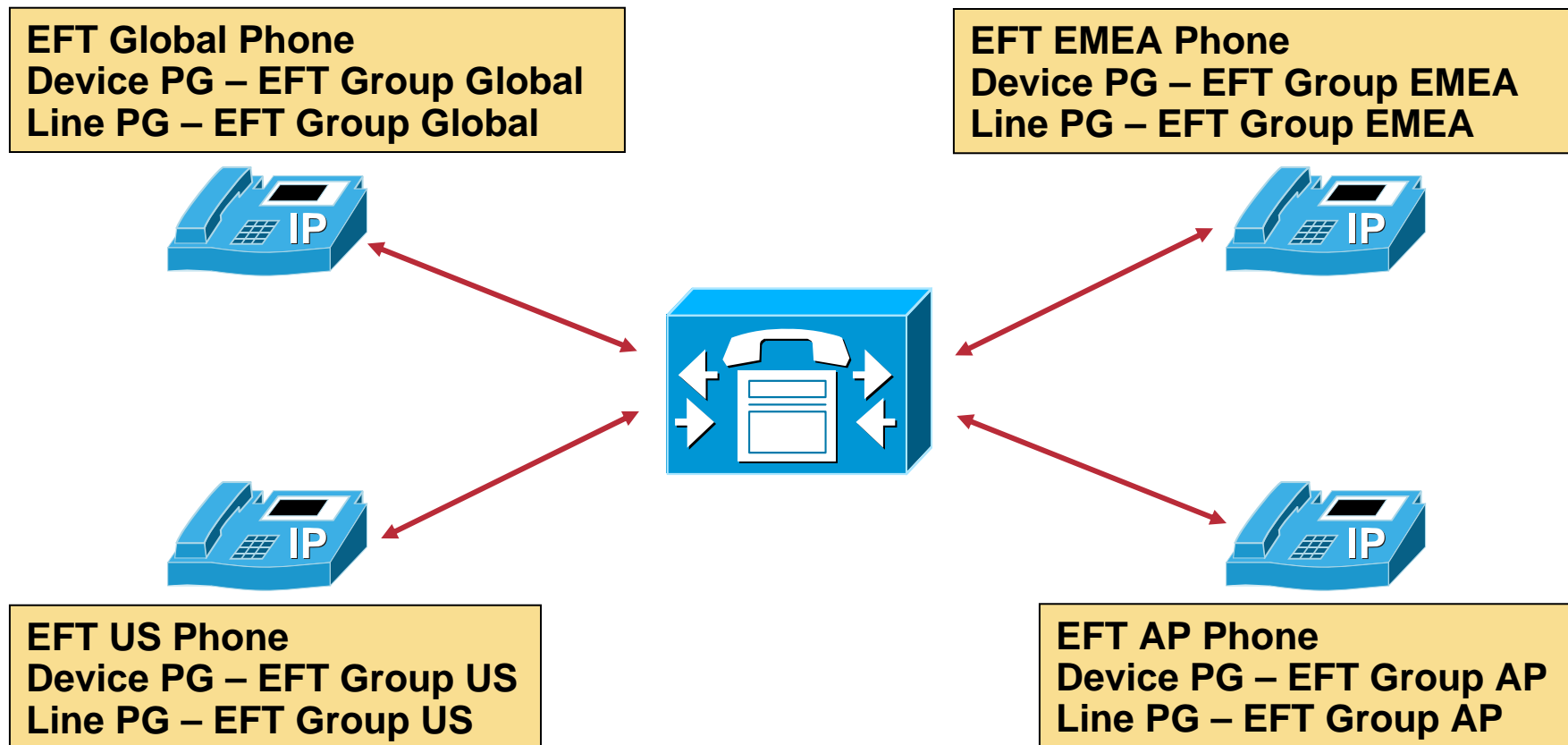
# Speed Dial BLF and Call List BLF

## Group based authorization

- **Group based mechanism to control which subscribers (watchers) can see which subscribers (notifiers).**
- **Defined by the Administrator.**
- **Each phone, trunk, and external watcher is a member of one particular group.**
- **Default Inter-Presence Group Permission is Deny (False)**

# Speed Dial BLF and Call List BLF

## Presence Group (PG) authorization example:



# Configure BLF

**To fully enable BLF on Cisco Unified CallManager 5.0 the following tasks need to be performed:**

- **Enable the BLF for Call Lists enterprise parameter.**
- **Allow subscriptions for presence cluster wide by using the Cisco Unified CallManager service parameters.**

# Configure BLF

The screenshot shows the Cisco CallManager Administration web interface. The browser title is "Enterprise Parameters Configuration - Mozilla Firefox". The address bar shows the URL "https://192.168.1.10/ccmadmin/serviceParamEdit.do?service=11". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The main content area is titled "Enterprise Parameters Configuration" and contains a table of parameters.

Parameter Name	Parameter Value	Suggested Value
<a href="#">Synchronization Between Auto Device Profile and Phone Configuration</a> *	True	True
<a href="#">Max Number of Device Level Trace</a> *	12	12
<a href="#">DSCP for Phone-based Services</a> *	default DSCP (000000)	de
<a href="#">DSCP for Phone Configuration</a> *	CS3(precedence 3) DSCP (011000)	CS
<a href="#">DSCP for Cisco CallManager to Device Interface</a> *	CS3(precedence 3) DSCP (011000)	CS3(pr... e 3) DSCP (0110
<a href="#">Connection Monitor Duration</a> *	120	12
<a href="#">Auto Registration Phone Protocol</a> *	SCCP	SCCP
<a href="#">BLF For Call Lists</a> *	Enabled	Disabled
<a href="#">TFTP Encrypted Configuration</a> *	False	False

A callout bubble with the text "Enable BLF for call lists" and a circled "1" points to the "BLF For Call Lists" parameter, which is highlighted with a red box. The current value is "Enabled" and the suggested value is "Disabled".

# Configure BLF

The screenshot shows the 'Service Parameter Configuration' web page in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10/ccmadmin/serviceParamEdit.do`. The page is divided into several sections:

- TLS Packet Capture Configurations:**
  - Packet Capture Enable \* (False)
  - Packet Capture Max File Size (MB) \* (2)
- Clusterwide Parameters (System - Presence):**
  - Presence Subscription Throttling Threshold \* (90000)
  - Presence Subscription Resume Threshold \* (80)
  - Default Inter-Presence Group Subscription \* (Allow Subscription) - This dropdown is highlighted with a red box and a callout bubble containing the text "Allow subscriptions between groups per default." and a circled "2".
- Clusterwide Parameters (System - Dual Mode Mobility):**
  - Integrated Dual-Mode Feature Enable \* (False)
  - H1 (Graceful) Handoff Number
  - H1 Handoff Number Partition (< None >)
  - H2 Handoff Number
  - H2 Handoff Number Partition (< None >)
  - Minimum Ring Timer \* (2)
  - Mobility Cisco CallManager Group (< None >)

At the bottom of the configuration area, there are three buttons: **Save**, **Set to Default**, and **Advanced**. Below the buttons is an information icon and the text: **i** - indicates required item.

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10".

# Configure Phones for BLF

**To allow phones to use BLF the following tasks need to be performed:**

- **Configure phone button templates using speed dial BLF.**
- **Assign templates to phones.**
- **Add BLF speed dials to phones.**
- **Configure subscription CSS on the phones.**

# Configure Phones for BLF

The screenshot shows the Cisco CallManager Administration web interface. The page title is "Phone Button Template Configuration". The breadcrumb navigation shows "Cisco CallManager Administration" > "Phone Button Template Configuration". The page is logged in as "CCMAdministrator".

The "Phone Button Template Information" section shows the "Button Template Name" as "7960-1line-4blf-1service SCCP".

The "Button Information" section contains a table with columns for "Button", "Feature", and "Label". A red box highlights the "Feature" column for buttons 2 through 9, where "Speed Dial BLF" is selected for buttons 2-5 and "Speed Dial" for button 6. A callout box with the text "Configure phone templates with Speed dial BLF" and a circled "1" points to this dropdown menu.

Button	Feature	Label
1	Line **	Line 1
2	Speed Dial BLF	Speed Dial BLF
3	Speed Dial BLF	Speed Dial BLF
4	Speed Dial BLF	Speed Dial BLF
5	Speed Dial BLF	Speed Dial BLF
6	Speed Dial	Service URL
7	Line	None
8	Privacy	None
9	Service URL	None



# Configure Phones for BLF

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL: `https://192.168.1.10/ccmadmin/phoneEdit.do?key=cb196e0f-1952-4ba9-b499-109a54c17eb5`. The page title is "Cisco CallManager Administration" and it is logged in as "CCMAdministrator".

The "Phone Configuration" section is active, showing the configuration for a Cisco 7960 phone. The "Association Information" table on the left lists several "Add a new BLF SD" options, with the first one highlighted by a red box and a callout bubble labeled "3" containing the text "Add new BLF speed dial".

The "Device Information" section on the right shows the phone's configuration. The "Phone Button Template\*" dropdown menu is highlighted with a red box and a callout bubble labeled "2" containing the text "Assign template to phone". The selected template is "7960-1line-4blf-1service SCCP".

Association Information	Phone Type	Device Information
1 Line [1] - 3003 in Chicago phones	Product Type: Cisco 7960	Registration: Registered with Cisco
2 Add a new BLF SD	Device Protocol: SCCP	IP Address: 10.128.192.51
3 Add a new BLF SD		MAC Address*: 000532D2F118
4 Add a new BLF SD		Description: Hardphone 1
5 Add a new BLF SD		Device Pool*: Chicago
6 Add a new SURF		Phone Button Template*: 7960-1line-4blf-1service SCCP
7 None		Softkey Template: < None >
8 None		Common Phone Profile*: Standard Common Phone Profile
9 None		Calling Search Space: < None >
10 None		AAR Calling Search Space: < None >
11 None		
12 None		
13 None		

# Configure Phones for BLF

**Select a DN from drop down menu or enter Destination** ④

**Enter a label that should be displayed on the phone** ⑤

	Destination	Directory Number	Label	Label ASCII
1		3001 in CHI-Phones	3001	3001
2	3007	< None >	Donna	Donna
3		< None >		
4		< None >		

	Destination	Directory Number	Label	Label ASCII
5		< None >		
6		< None >		
7		< None >		
8		< None >		
9		< None >		
10		< None >		
11		< None >		

# Configure Phones for BLF

Phone Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/phoneEdit.do?key=cb196e0f-1952-4ba9-b499-109a54c17eb5

33 None  
34 None  
----- Unassigned Associated Items -----  
35 Line [2] - Add a new DN  
36 Add a new SD  
37 Add a new SURL  
38 Add a new BLF SD  
39 Privacy  
40 None

**Protocol Specific Information**

Packet Capture Mode\* None  
Packet Capture Duration 0  
Presence Group\* Standard Presence group  
SCCP Phone Security Profile\* Standard SCCP Profile for Auto Registration  
SUBSCRIBE Calling Search Space **CHI-Intl-CSS**  
 Unattended Port  
 Require DTMF Reception  
 RFC2833 Disabled

**Expansion Module Information**

Module 1 < None >  
Module 1 Load Name  
Module 2 < None >  
Module 2 Load Name

**External Data Locations Information (Leave blank to use default)**

Information  
Authentication Server  
Proxy Server  
Idle

Done 192.168.1.10

**Select CSS to allow subscription of phones** 6

**Note: Only phones that can be reached by the subscription CSS can be monitored**

# Configure Presence Authorization

**To restrict the use of BLF the following tasks need to be performed:**

- **Configure presence group.**
- **Define relationship to other presence groups.**
- **Add phones to the presence group.**

# Configure Presence Authorization

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL `https://192.168.1.10/ccmadmin/presenceGroupFindList.do`. The page title is "Find and List Presence Groups". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The main content area is titled "Find and List Presence Groups" and includes a grid icon with a plus sign and a minus sign. Below this is a "Status" section indicating "1 records found". The "Search Options" section allows searching for a presence group by name, with a "Find" button and a checkbox for "Search Within Results".

The "Search Results" section displays a table with the following content:

Name	Copy
<a href="#">Standard Presence group</a>	

Below the table are buttons for "Add New", "Select All", "Clear All", and "Delete Selected", along with a "Rows per Page" dropdown set to 50. A yellow callout bubble with the text "Add a new group" and a circled "1" points to the "Add New" button.

# Configure Presence Authorization

The screenshot shows the Cisco CallManager Administration web interface for configuring a presence group. The browser window is titled "Presence Group Configuration - Mozilla Firefox" and the URL is "https://192.168.1.10/ccmadmin/presenceGroupEdit.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator".

Five numbered callouts highlight key configuration steps:

- 1**: A red box highlights the "Save" button at the bottom left of the form.
- 2**: A yellow callout points to the "Name\*" and "Description" input fields, which contain "CHI-Phones" and "All phones CHI" respectively.
- 3**: A yellow callout points to the "Modify Relationship to Other Presence Groups" section, specifically the dropdown menu showing "Standard Presence group".
- 4**: A yellow callout points to the "Subscription" dropdown menu, which is open to show options: "Use System Default", "Allow Subscription", and "Disallow Subscription".
- 5**: A yellow callout points to the "Save settings" text, which is positioned above the "Save" button.

Other visible elements include a "Status" section showing "Ready", a "Related Links" section with "Back To Find/List", and a "Save" button at the bottom left. A footer note states "\* - indicates required item." The status bar at the bottom shows "Done" and the IP address "192.168.1.10".

# Configure Presence Authorization

Phone Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10/ccmadmin/phoneEdit.do?key=cb196e0f-1952-4ba9-b499-109a54c17eb5

33 None  
34 None  
----- Unassigned Associated Items -----  
35 Line [2] - Add a new DN  
36 Add a new SD  
37 Add a new SURL  
38 Add a new BLF SD  
39 Privacy  
40 None

**Protocol Specific Information**

Packet Capture Mode\* None  
Packet Capture Duration 0  
Presence Group\* Standard Presence group  
SCCP Phone Security Profile\* CHI-Phones  
SUBSCRIBE Calling Search Space CHI-MTS

Unattended Port  
 Require DTMF Reception  
 RFC2833 Disabled

**Expansion Module Information**

Module 1  
Module 1 Load Name  
Module 2 < None >  
Module 2 Load Name

**External Data Locations Information (Leave blank to use default)**

Information  
Directory  
Messages  
Services  
Authentication Server  
Proxy Server  
Idle

Done 192.168.1.10

**Add phone to presence group** 6

# Summary

- **Speed dial BLF and call list BLF allow to monitor the status of other DNs within the network.**
- **To use BLF special system values need to be defined.**
- **To use BLF phones need to be configured using a BLF enabled phone button template and a subscription CSS.**
- **To restrict BLF throughout the cluster presence authorization needs to be configured.**



# CISCO SYSTEMS





# Cisco Unified CallManager 5.0 Features and Applications

## Configuring Telephony Features

# Objectives

- **Configuring Call Park**
- **Configuring Call Pickup**
- **Configuring Conferencing**

# Configuring Call Park

## Call Park configuration steps:

1. Go to Call Routing > Call Park.
2. Add new Call Park range.
3. Specify a Call Park DN range.

# Configuring Call Park

The screenshot shows the Cisco CallManager Administration web interface. The browser window title is "Cisco CallManager Console - Mozilla Firefox". The address bar shows the URL "https://192.168.1.10:8443/ccmadmin/showHome.do". The page header includes "Cisco CallManager Administration" and "Logged in as: CCMAdministrator". A navigation menu is visible with options like "System", "Call Routing", "Media Resources", etc. The "Call Routing" menu is expanded, showing a list of options including "AAR Group", "Dial Rules", "Route Filter", "Route/Hunt", "SIP Route Pattern", "Class of Control", "Client Matter Codes", "Forced Authorization Codes", "Translation Pattern", "Call Park", "Call Pickup Group", "Directory Number", "Meet-Me Number/Pattern", "Route Plan Report", and "Dial Plan Installer". The "Call Park" option is highlighted. A yellow callout box with a circled "1" points to "Call Park" and contains the text "Go to Call Routing > Call Park". The bottom of the browser window shows the URL "https://192.168.1.10:8443/ccmadmin/callParkFindList.do" and the IP address "192.168.1.10:8443".

# Configuring Call Park

The screenshot shows a Mozilla Firefox browser window displaying the Cisco CallManager Administration interface. The address bar shows the URL `https://192.168.1.10:8443/ccmadmin/callParkFindList.do`. The page title is "Find and List Call Park Numbers". The interface includes a navigation menu with options like System, Call Routing, Media Resources, Voice Mail, Device, Application, User Management, Bulk Administration, and Help. A "Log Off" button is visible in the top right. The main content area has a "Search Options" section with a text input field for "Find Call Park Numbers where" and a "Find" button. Below this is a "Search Results" section with the message "No active query. Please enter your search criteria using the options above." and an "Add New" button. A yellow callout box with a circled "2" points to the "Add New" button, containing the text "Add a new Call Park range."

# Configuring Call Park

Call Park Number Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccmadmin/callParkEdit.do

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as:CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management **3** Call Administration Help Log Off

Call Park Number Configuration Find/List Go

Status

Status: Ready

Call Park Configuration.

Call Park Number/Range\* 29XX

Description SJC Call Park

Partition SJC-Phones

Cisco CallManager\* CM\_CM1

Save

\* - indicates required item.

Done 192.168.1.10:8443

**Specify Call Park DN range including pattern, partition and CallManager**

# Configuring Call Pickup

## Call Pickup configuration steps:

1. Go to Call Routing > Call Pickup.
2. Add a new Call Pickup group.
3. Configure a Call Pickup group.
4. Assign Call Pickup group to multiple lines.



# Configuring Call Pickup

The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser. The browser's address bar displays the URL `https://192.168.1.10:8443/ccmadmin/showHome.do`. The page header includes the title "Cisco CallManager Administration" and the user "Logged in as: CCMAdministrator". A navigation menu is visible at the top, with "Call Routing" selected. A dropdown menu is open under "Call Routing", listing various options. The "Call Pickup Group" option is highlighted with a blue background. A yellow callout box with a circled "1" contains the text "Go to CallManager Administration" and "Go to Call Routing > Call Pickup Group". The background of the page shows a blurred image of a person working at a computer.

Navigation Cisco CallManager Administration Go

Cisco CallManager Administration For Cisco IP Telecommunication Solutions Logged in as: CCMAdministrator

System Call Routing Media Resources Voice Mail Device Application User Management Bulk Administration Help Log Off

AAR Group  
Dial Rules  
Route Filter  
Route/Hunt  
SIP Route Pattern  
Class of Control  
Client Matter Codes  
Forced Authorization Codes  
Translation Pattern  
Call Park  
Call Pickup Group  
Directory Number  
Meet-Me Number/Pattern  
Route Plan Report  
Dial Plan Installer

1 Go to CallManager Administration  
Go to Call Routing >  
Call Pickup Group

https://192.168.1.10:8443/ccmadmin/callPickupGroupFindList.do 192.168.1.10:8443

# Configuring Call Pickup

The screenshot shows a Mozilla Firefox browser window displaying the Cisco CallManager Administration interface. The page title is "Find and List Call Pickup Groups". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The main content area is titled "Find and List Call Pickup Groups" and features a search interface. A yellow callout box with a circled "2" and the text "Click Add New to add new group." points to the "Add New" button located below the search results section.

**Search Options**  
Find Call Pickup Group where [Call Pickup Group Name] [Begin/End Time] [Find]  Search Within Results

**Search Results**  
No active query. Please enter your search criteria using the options above.  
[Add New](#)

# Configuring Call Pickup

The screenshot shows the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar displays `https://192.168.1.10:8443/ccadmin/callPickupGroupEdit.do`. The page title is "Call Pickup Group Configuration". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The main content area is titled "Call Pickup Group Configuration" and includes a "Related Links" section with a "Back To Find/List" button. A yellow callout box with the number "3" and the text "Specify group settings." points to the "Call Pickup Group Information" section. This section contains the following fields:

- Call Pickup Group Name\*: Group1 San Jose
- Call Pickup Group Number\*: 2801
- Partition: SJC-Phones

Below this section is the "Associated Call Pickup Group Information" section, which includes a "Find Pickup Numbers by Numbers/Partition" search area. The search results show "(No Matches Found)".

The status bar at the bottom of the browser window shows "Done" and the IP address "192.168.1.10:8443".

# Configuring Call Pickup

Directory Number Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccadmin/directoryNumberEdit.do?key=66d90053-935f-ec72-c061-479153551f8a&mapkey=74e26334-46a1-73c7-0d57-c8

Auto Answer\* Auto Answer Off

### Call Forward and Call Pickup Settings

	Voice Mail Destination	Calling Search Space
Forward All	<input type="checkbox"/> or	< None >
Secondary Calling Search Space for Forward All		< None > Find
Forward Busy Internal	<input type="checkbox"/> or	< None >
Forward Busy External	<input type="checkbox"/> or	< None >
Forward No Answer Internal	<input type="checkbox"/> or	< None >
Forward No Answer External	<input type="checkbox"/> or	< None >
Forward No Coverage Internal	<input type="checkbox"/> or	< None >
Forward No Coverage External	<input type="checkbox"/> or	< None >
Forward on CTI Failure	<input type="checkbox"/> or	< None >

No Answer Ring Duration (seconds)

Call Pickup Group Group1 San Jose in SJC-Phones

### MLPP Alternate Party Settings

Target (Destination)

MLPP Calling Search Space < None >

MLPP No Answer Ring Duration (seconds)

### Line 1 on Device SEP00123F19CBD6

Display (Internal) Display text for a line appearance is intended for displaying text such as a

Find: pick Find Next Find Previous Highlight Match case

Done 192.168.1.10:8443

# Configuring Conferencing

## Conferencing configuration steps:

1. Ensure that the IP Voice Media Stream App service is running.
2. Go to Media Resources > Conference Bridge.
3. Ensure that a software Conference Bridge is registered.
4. Configure Conference Bridge settings.
5. Tweak IP Media Streaming App service parameters. (optional).
6. Tweak CallManager service parameters (optional).

# Configuring Conferencing

The screenshot shows the Cisco CallManager Serviceability interface. The browser title is "Cisco CallManager Serviceability-Service Activation - Mozilla Firefox". The address bar shows the URL: <https://192.168.1.10:8443/ccmservice/serviceactivation.jsp?htxtNodeID=8fb74274-671f-491a-93e3-afb5ee74ed73&htxtSubmit=Status&publisherNodeID=8fb74274-671f-491a-93e3-afb5ee74ed73>. The page title is "Cisco CallManager Serviceability For Cisco IP Telecommunication Solutions".

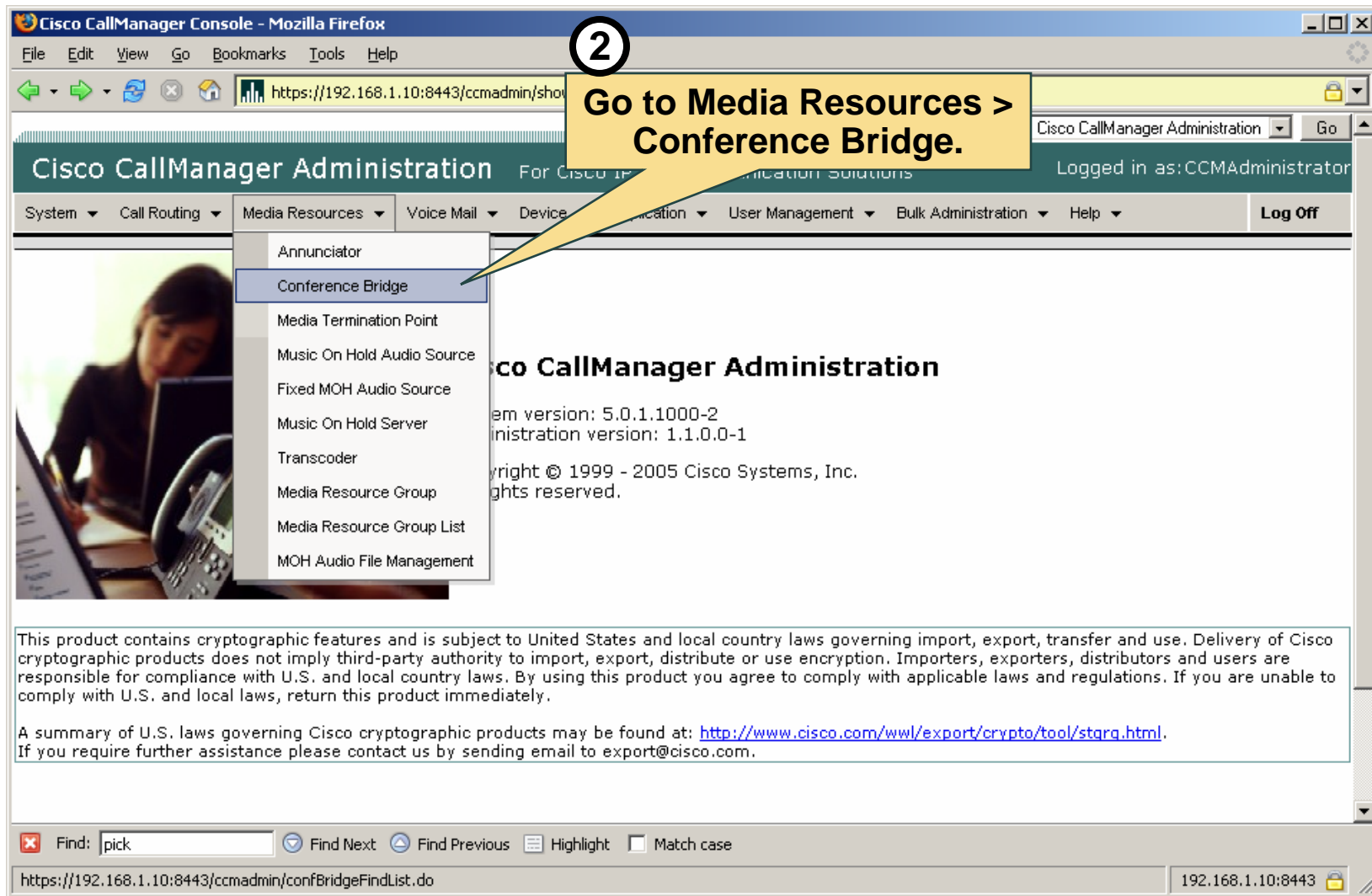
The "Service Activation" section shows the status of various services. The "Status" is "Ready". The "Select Server" dropdown is set to "192.168.1.10".

The "CM Services" table shows the following services and their status:

Service Name	Status
<input checked="" type="checkbox"/> Cisco CallManager	Activated
<input checked="" type="checkbox"/> Cisco Tftp	Activated
<input type="checkbox"/> Cisco Messaging Interface	Deactivated
<input checked="" type="checkbox"/> Cisco IP Voice Media Streaming App	Activated
<input checked="" type="checkbox"/> Cisco CTIManager	Activated
<input type="checkbox"/> Cisco CallManager Attendant Console Server	Deactivated
<input checked="" type="checkbox"/> Cisco Extension Mobility	Activated
<input checked="" type="checkbox"/> Cisco Extended Functions	Activated
<input checked="" type="checkbox"/> Cisco CallManager Cisco IP Phone Services	Activated

A yellow callout box with a circled "1" points to the "Cisco IP Voice Media Streaming App" row, with the text "Verify service is running." The status of this service is "Activated".

# Configuring Conferencing



The screenshot shows the Cisco CallManager Administration web interface in a Mozilla Firefox browser. The browser's address bar displays the URL `https://192.168.1.10:8443/ccmadmin/show`. The page title is "Cisco CallManager Administration" and it indicates the user is logged in as "CCMAdministrator". A navigation menu at the top includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The "Media Resources" menu is expanded, showing a list of options: "Annunciator", "Conference Bridge", "Media Termination Point", "Music On Hold Audio Source", "Fixed MOH Audio Source", "Music On Hold Server", "Transcoder", "Media Resource Group", "Media Resource Group List", and "MOH Audio File Management". The "Conference Bridge" option is highlighted. A yellow callout box with a circled "2" contains the text "Go to Media Resources > Conference Bridge." Below the navigation menu, the main content area displays "Cisco CallManager Administration" and system information: "System version: 5.0.1.1000-2" and "Administration version: 1.1.0.0-1". A copyright notice for Cisco Systems, Inc. is also visible. At the bottom of the page, there is a search bar with the text "Find: pick" and buttons for "Find Next", "Find Previous", "Highlight", and "Match case". The browser's status bar shows the current page URL as `https://192.168.1.10:8443/ccmadmin/confBridgeFindList.do` and the IP address `192.168.1.10:8443`.

# Configuring Conferencing

The screenshot shows the Cisco CallManager Administration web interface. The page title is "Find and List Conference Bridges". The navigation menu includes "System", "Call Routing", "Media Resources", "Voice Mail", "Device", "Application", "User Management", "Bulk Administration", and "Help". The user is logged in as "CCMAdministrator".

The "Find and List Conference Bridges" section shows a status of "1 records found". The search options include a search box and a "Find" button. The search results are displayed in a table:

Conference Bridge Name	Description	Device Pool	Status	IP Address	Copy
<a href="#">CFB_2</a>	CFB_CM1	<a href="#">SanJose</a>	Registered with 192.168.1.10	192.168.1.10	Not Allowed

A callout box with the number "3" and the text "Verify Bridge is registered." points to the "CFB\_2" entry in the table. Below the table are buttons for "Add New", "Select All", "Clear All", "Delete Selected", and "Reset Selected", along with a "Rows per Page" dropdown set to 50.



# Configuring Conferencing

The screenshot displays the Cisco CallManager Administration web interface in Mozilla Firefox. The browser address bar shows the URL: `https://192.168.1.10:8443/ccmadmin/confBridgeEdit.do?key=9fc7ec46-6c2f-47f0-8032-c7e34e3f3ad1`. The page title is "Conference Bridge Configuration".

The main content area shows the configuration for a Conference Bridge named "CFB\_2 (CFB\_CM1)". The "Software Conference Bridge Info" section is highlighted with a red box and a callout bubble containing the number "4" and the text "Configure Conference Bridge settings." The fields in this section are:

- Conference Bridge Type\*: Cisco Conference Bridge Software
- Host Server: 192.168.1.10
- Conference Bridge Name\*: CFB\_2
- Description: CFB\_CM1
- Device Pool\*: SanJose
- Location\*: SJC

Buttons for "Save" and "Reset" are visible below the form. A legend at the bottom left indicates that an asterisk (\*) denotes a required item. The browser's search bar at the bottom shows the word "pick" and various search options like "Find Next", "Find Previous", "Highlight", and "Match case".

# Configuring Conferencing

The screenshot displays the 'Service Parameter Configuration' web interface in Mozilla Firefox. The URL is <https://192.168.1.10:8443/ccadmin/serviceParamEdit.do?server=8fb74274-671f-491a-93e3-afb5ee74ed73&service=3>. The interface is divided into several sections:

- Annunciator (ANN) Parameters:** Call Count \* (48), Run Flag \* (True).
- Conference Bridge (CFB) Parameters:** Call Count \* (48), Run Flag \* (True). This section is highlighted with a red box.
- Media Termination Point (MTP) Parameters:** Call Count \* (48), Run Flag \* (True).
- Clusterwide Parameters (Parameters that apply to all servers):** Supported MOH Codecs \* (711 mulaw), MOH Fixed Audio Quality level \* (Medium Quality), IP DSCP to Cisco CallManager \* (CS3(precedence 3) DSCP (011000)), Multicast MOH IP DSCP \* (EF DSCP (101110)), MTP DTMF Duration \* (100), MTP DTMF Power (volume) \* (9).

A yellow callout box with a circled '5' points to the Call Count and Run Flag parameters in the Conference Bridge section, containing the text: "Tune Call Count and Run Flag parameters (optional)".

At the bottom, there is a search bar with the text 'pick' and a 'Done' status. The status bar at the bottom right shows the IP address 192.168.1.10:8443.

# Configuring Conferencing

Service Parameter Configuration - Mozilla Firefox

File Edit View Go Bookmarks Tools Help

https://192.168.1.10:8443/ccadmin/serviceParamEdit.do?server=8fb74274-671f-491a-93e3-afb5ee74ed73&service=0

Call Back Notification Audio File Name *	CallBack.raw	CallBack.raw
Connection Proposal Type *	Connection Retention	Connection Retention
Connection Response Type *	Default to Connection Retention	Default to Connection Retention
Call Back Request Protection T1 Timer *	10	10
Call Back Recall T3 Timer *	20	20
Call Back Calling Search Space	< None >	
No Path Reservation *	True	True
Set Private Numbering Plan for Call Back *	False	False

**Clusterwide Parameters (Route Plan)**

Stop Routing on Out of Bandwidth Flag *	False	False
Stop Routing on Unallocated Number Flag *	True	True
Stop Routing on User Busy Flag *	True	True

**Clusterwide Parameters (Service)**

Default Network Hold MOH Audio Source ID *	1	1
Default User Hold MOH Audio Source ID *	1	1
Duplex Streaming Enabled *	False	False
Maximum Ad Hoc Conference *	4	4
Maximum MeetMe Conference Unicast *	4	4
Media Exchange Interface Capability Timer *	8	8

Done

192.168.1.10:8443

**6**

**Tune the maximum number of participants.**

# Summary

- **Add Call Park patterns to a partitions.**
- **Configure call pick up groups and assign them to phones.**
- **AdHoc conferencing is done via the conference bridge.**

# CISCO SYSTEMS

