Understanding Cisco Enterprise IP Communication

Introducing Cisco Enterprise IP Communication Components

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Objectives

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

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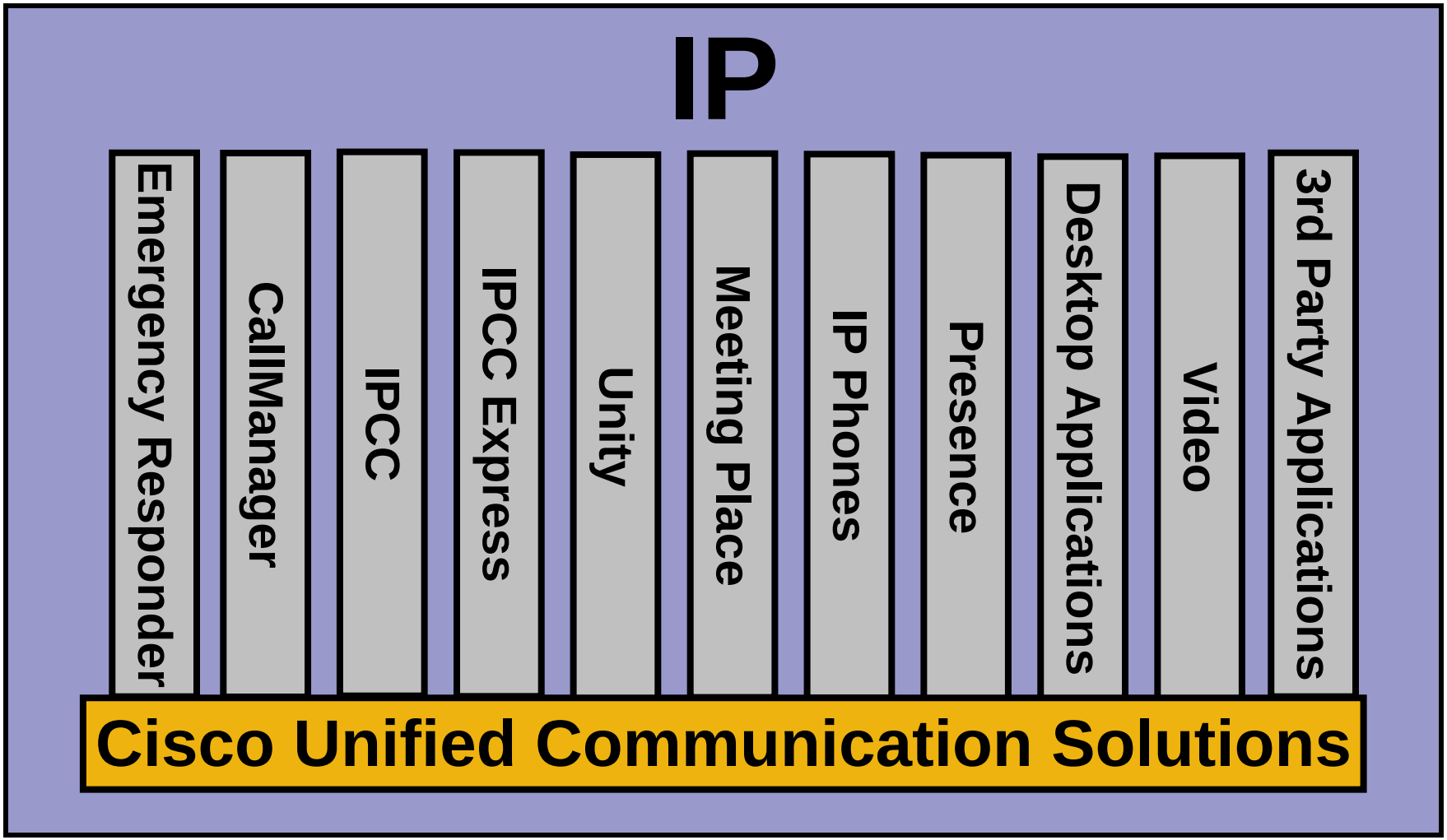
Cisco Unified Communications Overview



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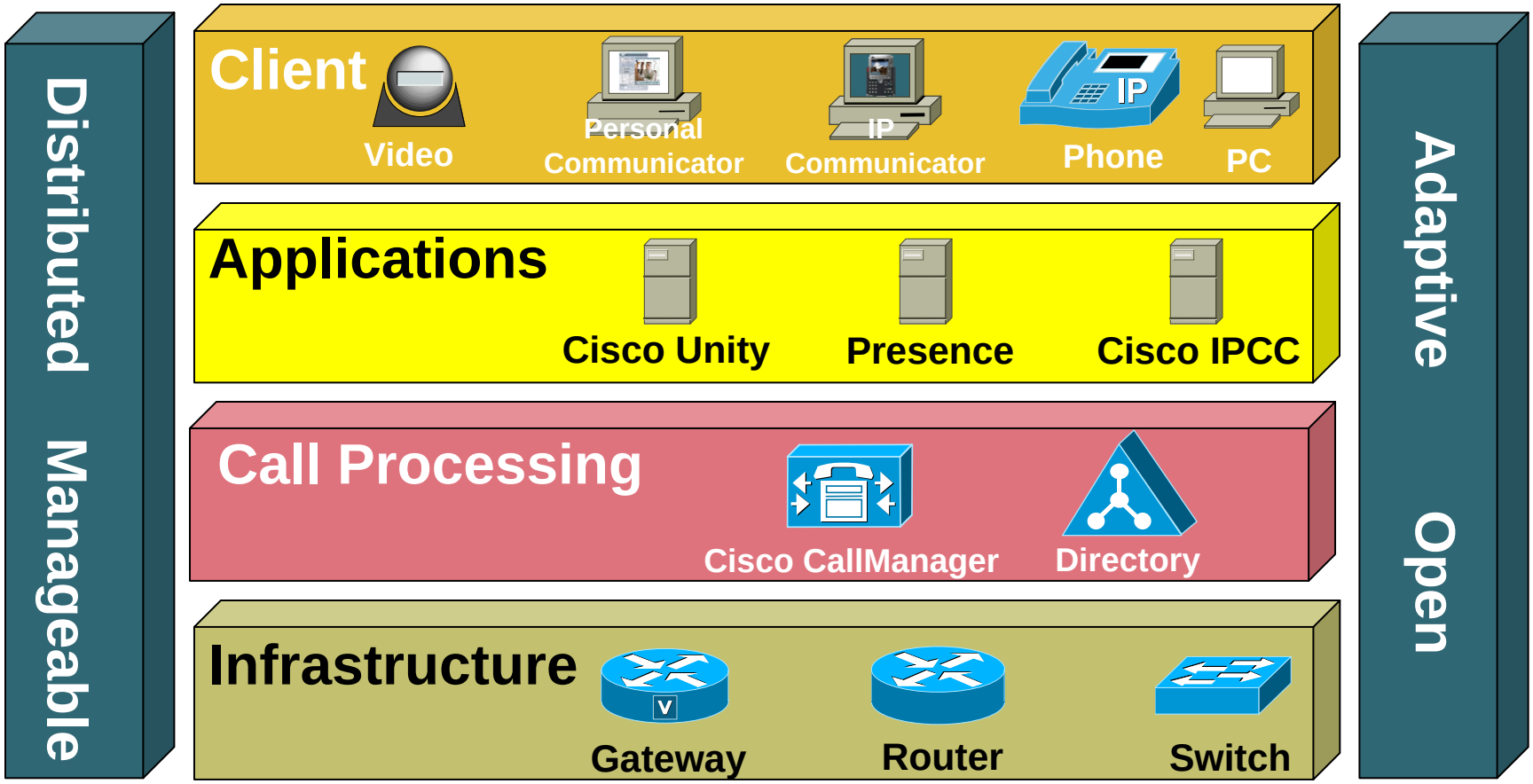
Cisco Unified Communication Layers



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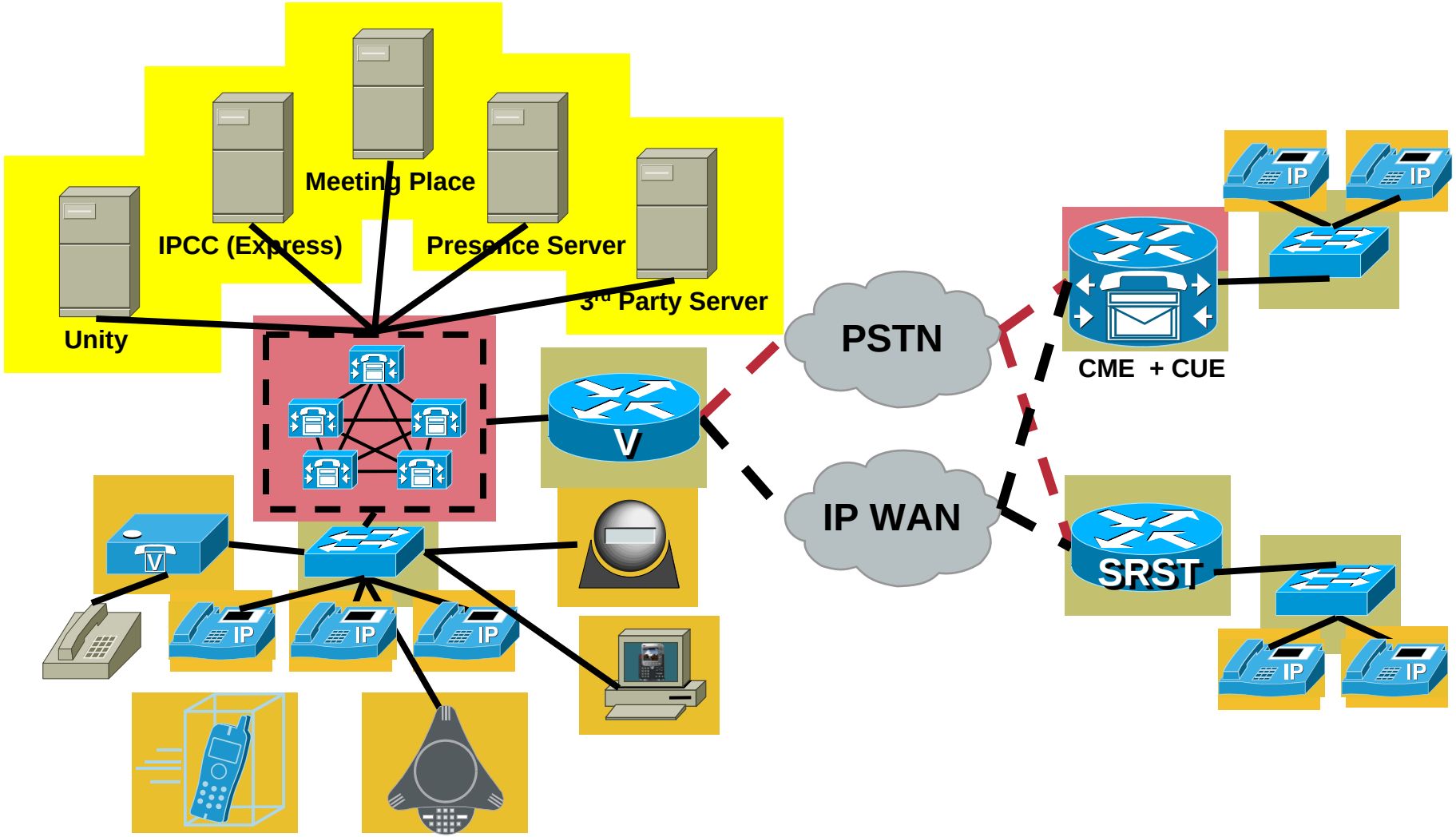
Cisco IP Communication Components



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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.**

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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.

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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

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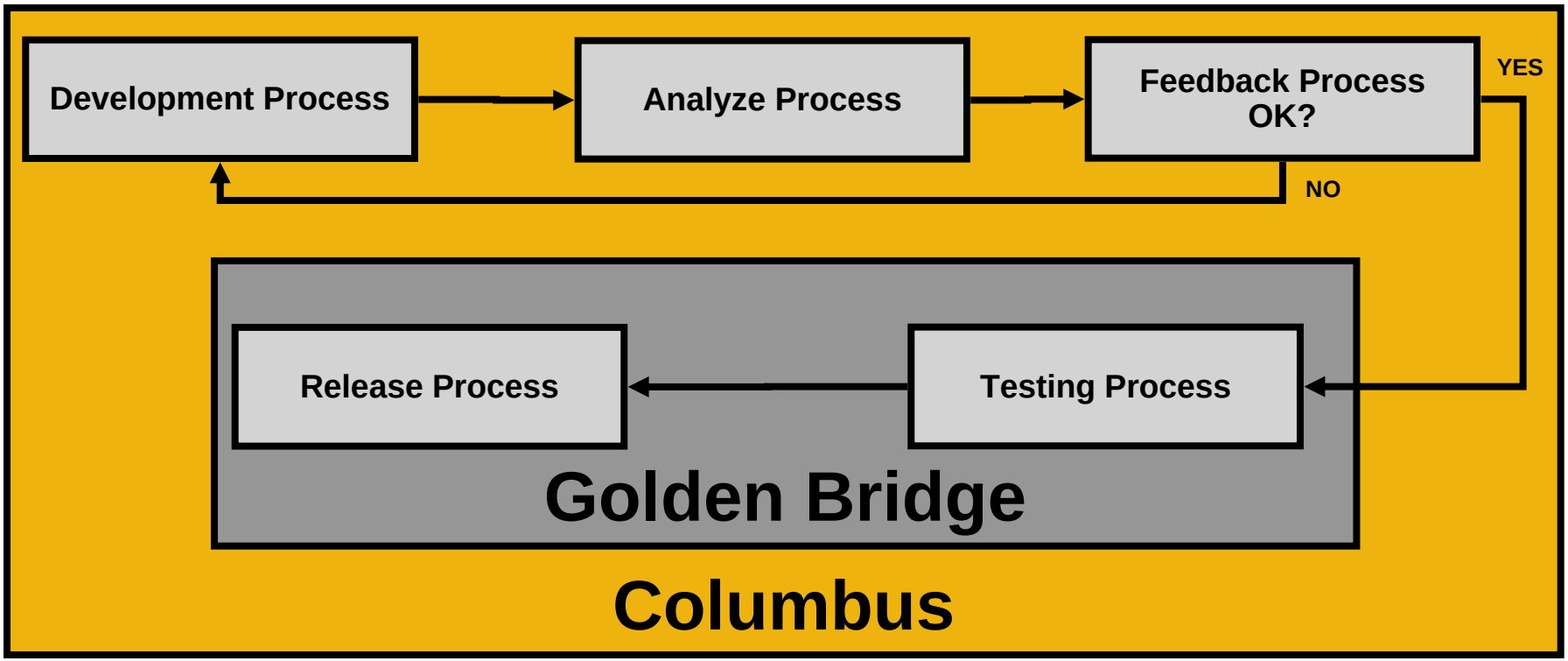
Columbus Testing, Design, and Implementation Process

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



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Columbus versus Golden Bridge

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

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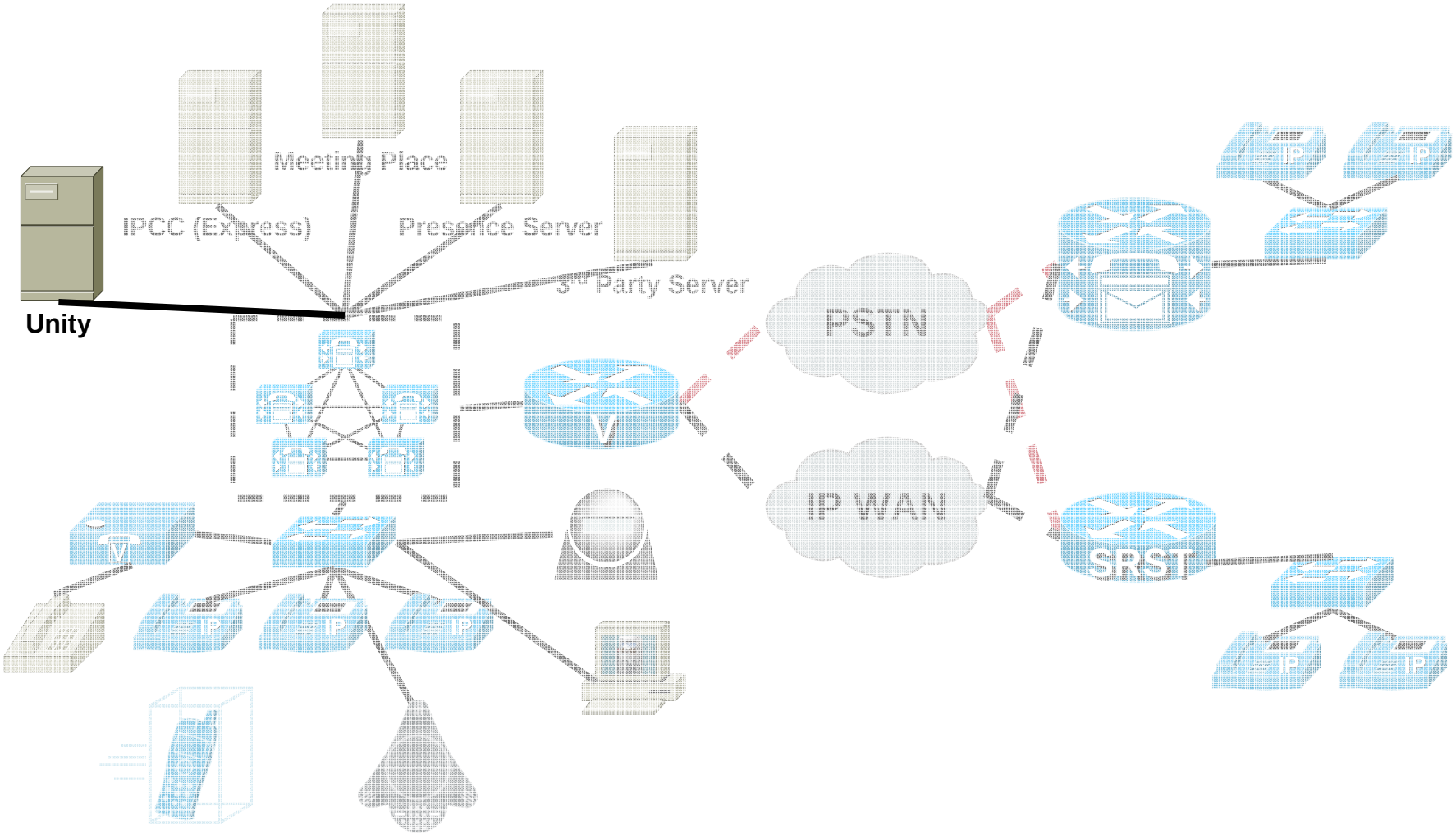
Cisco Unity



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Cisco Unity

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

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

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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.**

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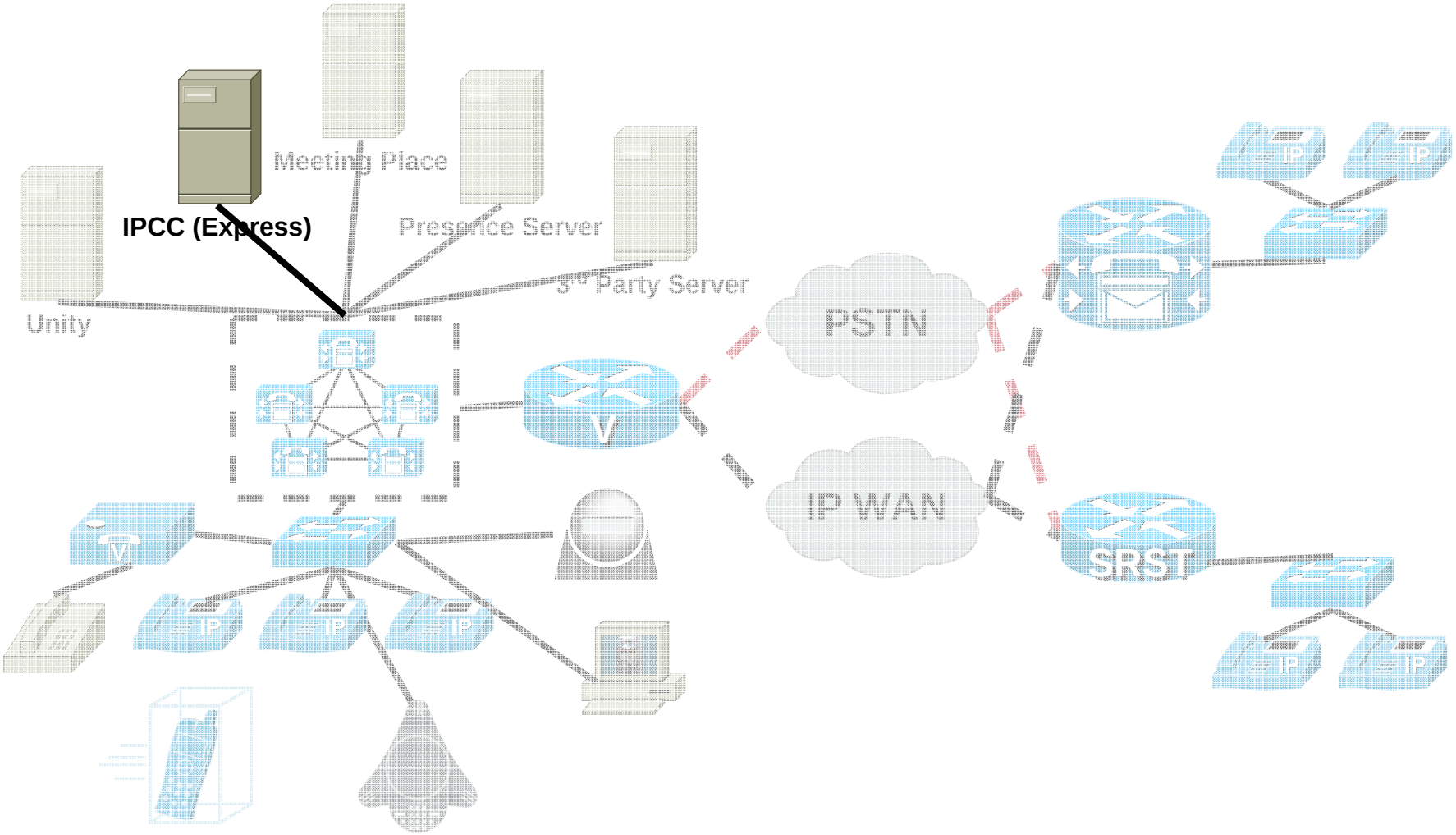
Cisco IP Contact Center



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Cisco Contact Center Vision

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

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IPCC Enterprise versus IPCC Express

Criteria	Cisco IPCC Express Edition 4.0	Cisco IPCC Enterprise Edition 7.0
Virtual Contact Center in *Centralized* Call Deployment Model (One CallManager Cluster)	✔	✔
Virtual Contact Center in *Distributed* Call Deployment Model (Multiple CallManager Clusters)	⊘	✔
Integrated Multi-Channel	⊘	✔
Blended Inbound/Outbound	⊘	✔
Scalability > IPCCX to 300 agents, IPCCE to thousands	⊘	✔
Redundancy (NOTE: IPCC Enterprise has superior redundancy to that provided by IPCC Express e.g. preservation of calls/data in queue)	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	✔	⊘

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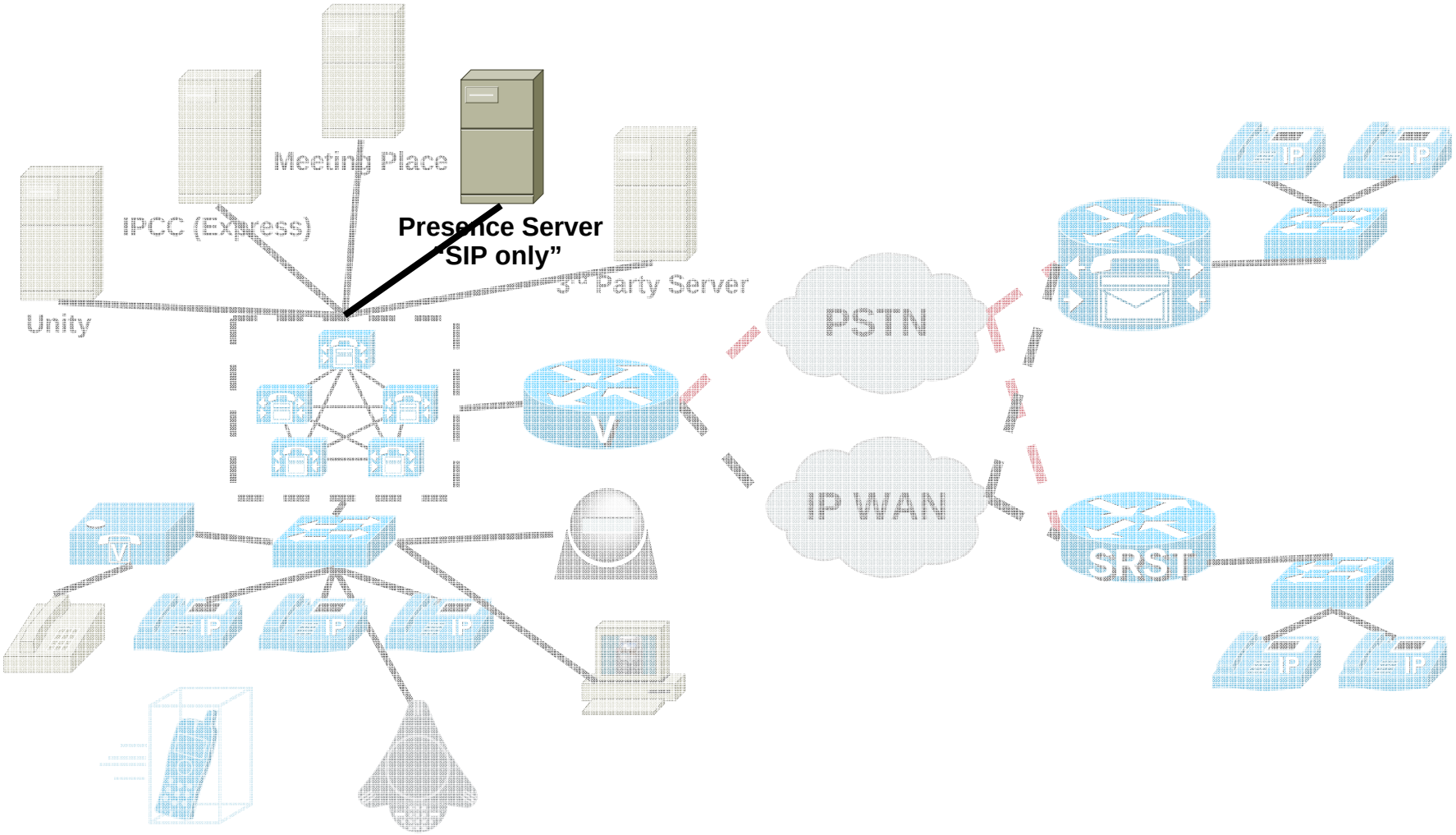
Cisco Unified Presence Server



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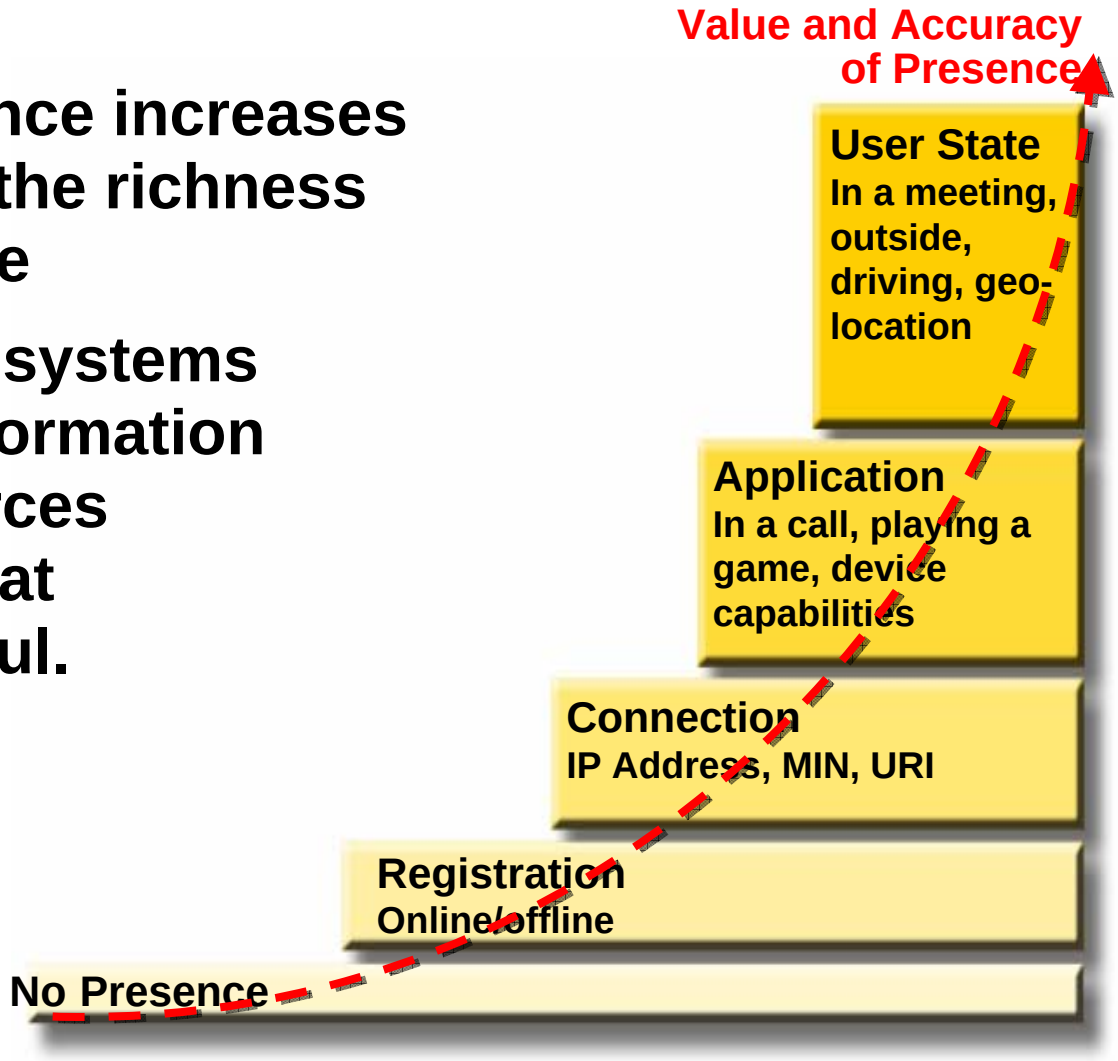
Presence Overview: Introduction

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.



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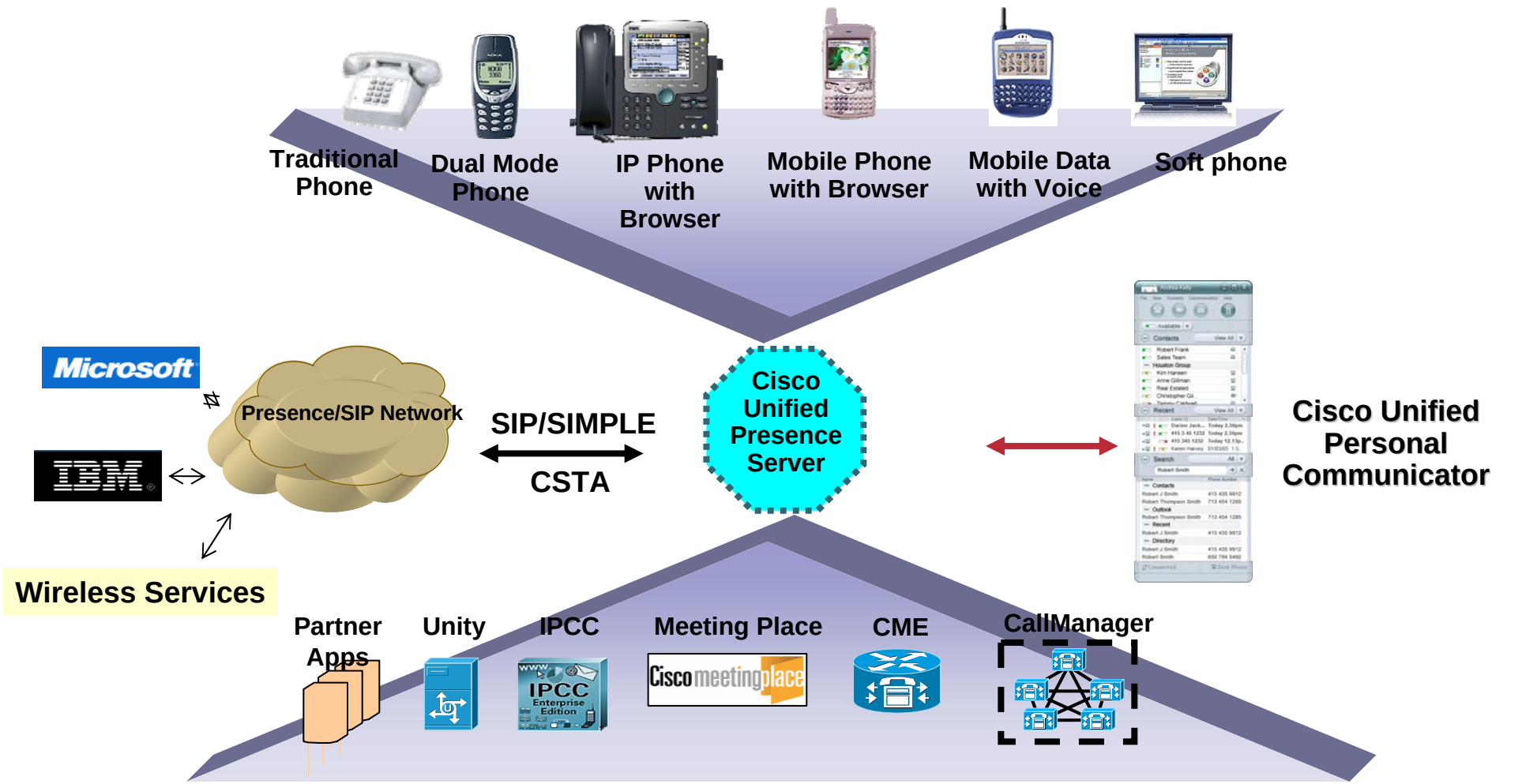
Cisco Unified Presence Server: Overview



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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**

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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**

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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.

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Understanding Cisco Enterprise IP Communication

Understanding Cisco Unified CallManager Concepts

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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

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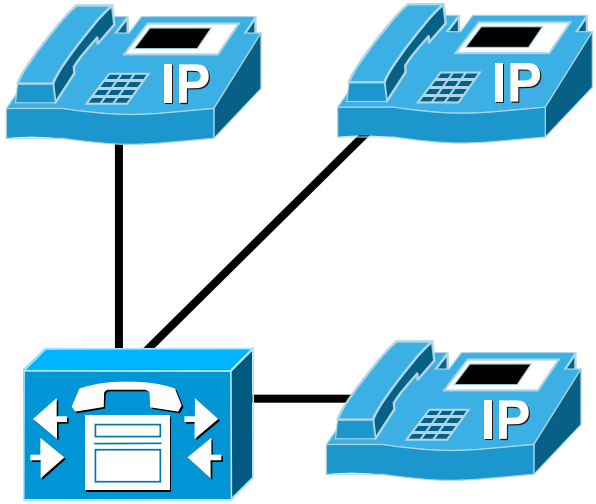
Cisco Unified CallManager Functions Overview

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**



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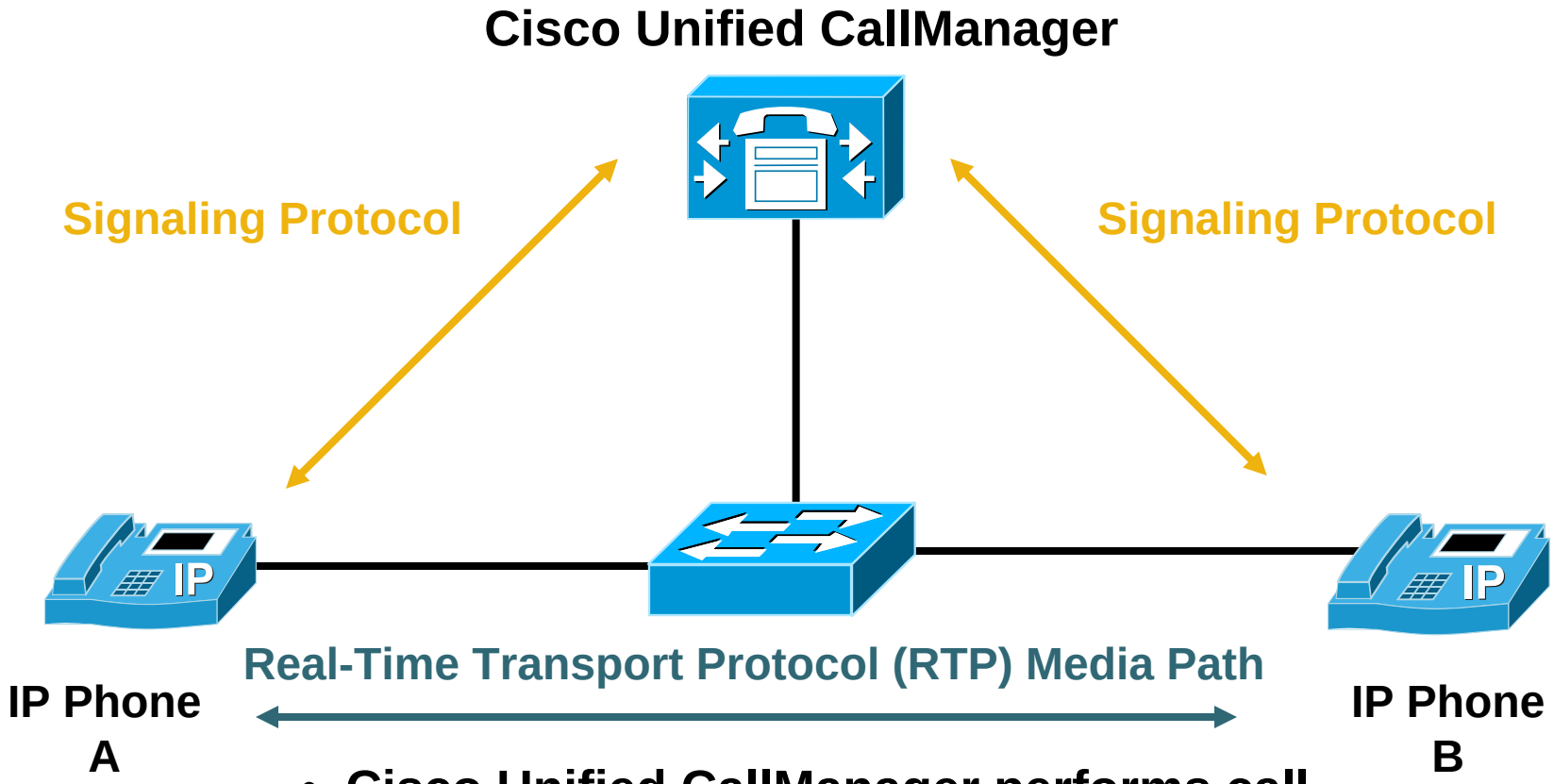
Cisco Unified CallManager Functions (Cont.)



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

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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

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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

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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

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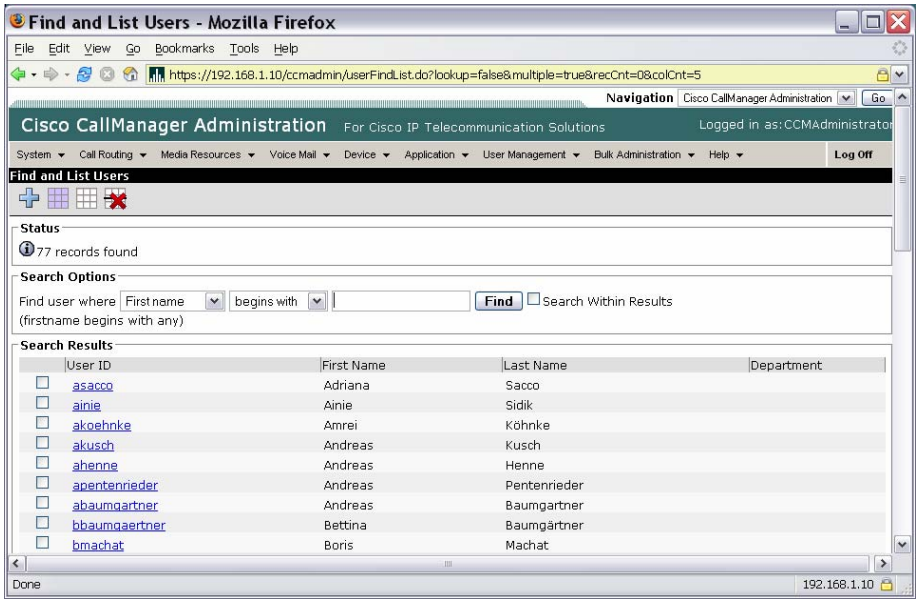
Administration Enhancements

- **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**



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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

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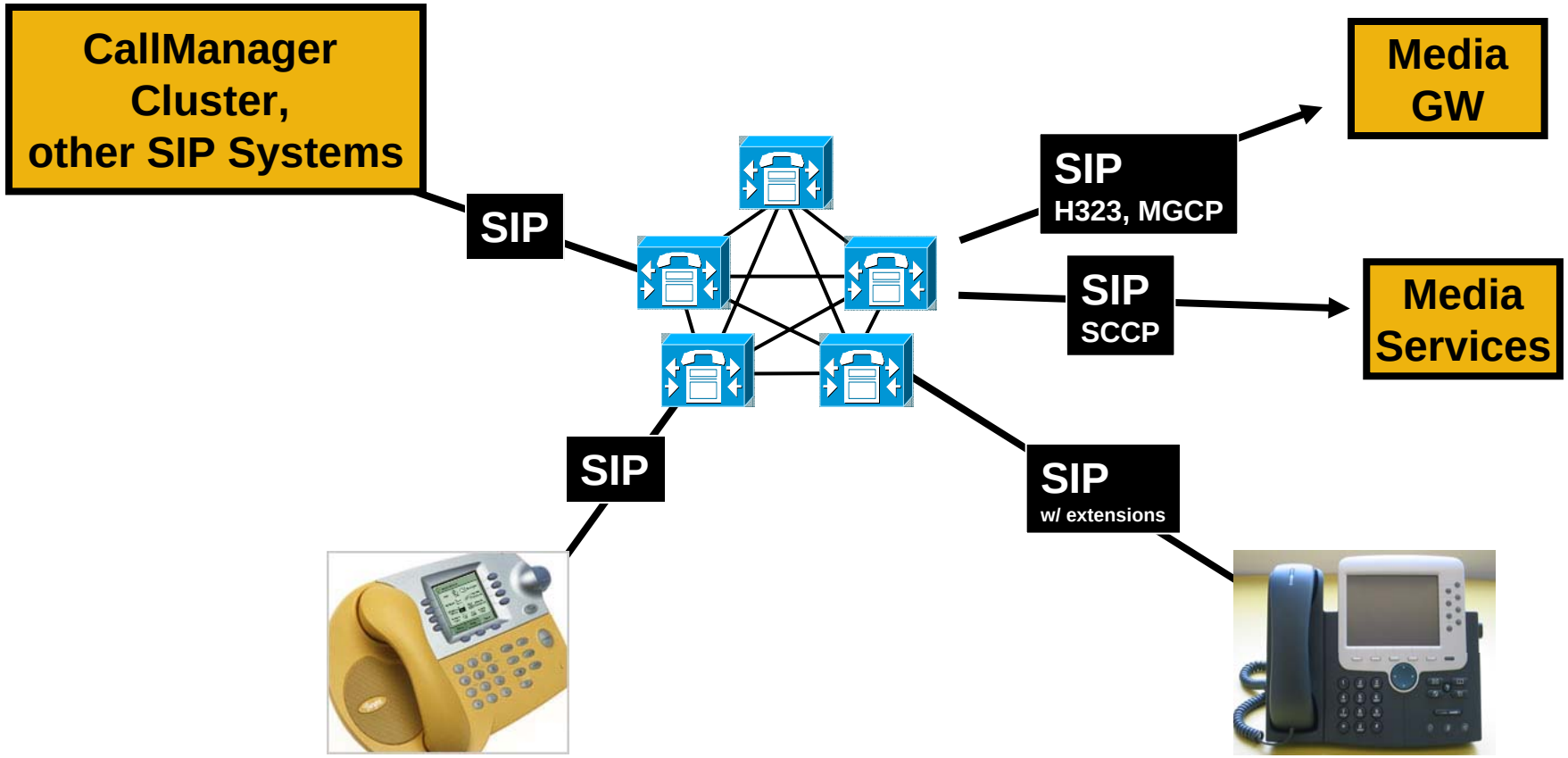
SIP Possibilities Overview CCM 5.0



SIP RFC 3261 Compliant Phones

Next-Gen Cisco SIP Phones

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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

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Multi-Byte Character Support (Japanese)

Incoming Call

Corporate Directory

Speed Dials

Directories

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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.**

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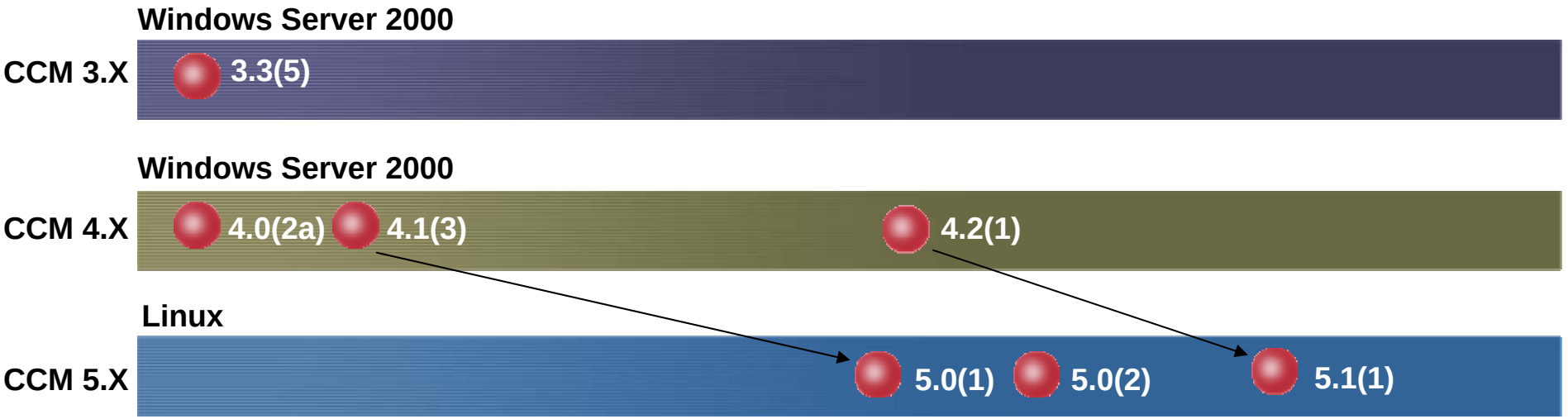
Contrast
Cisco Unified CallManager 4.x and 5.0



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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

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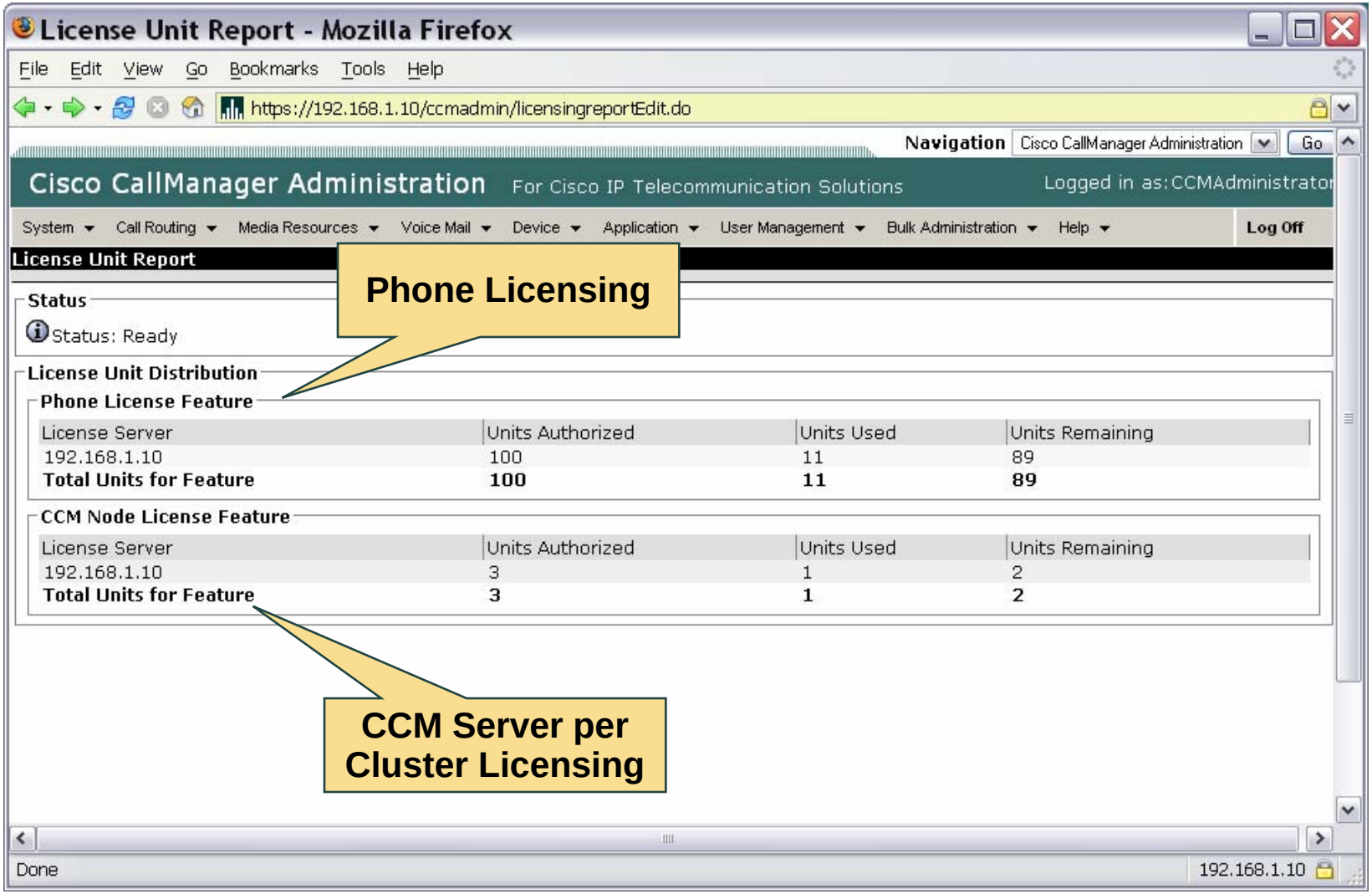
License Unit Report



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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.

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Understanding Cisco Enterprise IP Communication

Evaluating Cisco Unified CallManager Deployment Models

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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

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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.**

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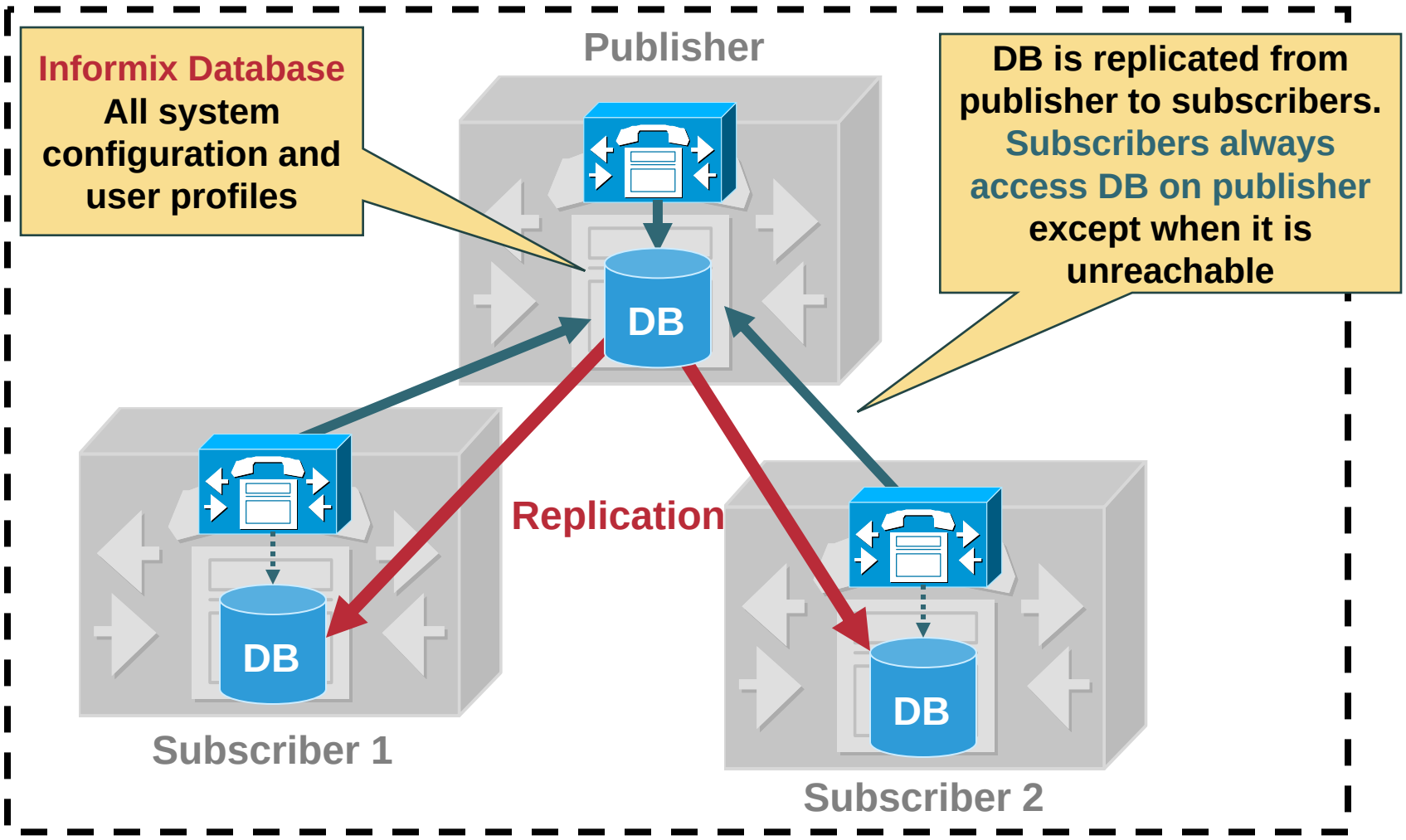
CallManager 5.0 Directory Architecture



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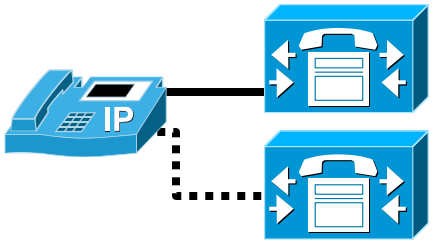
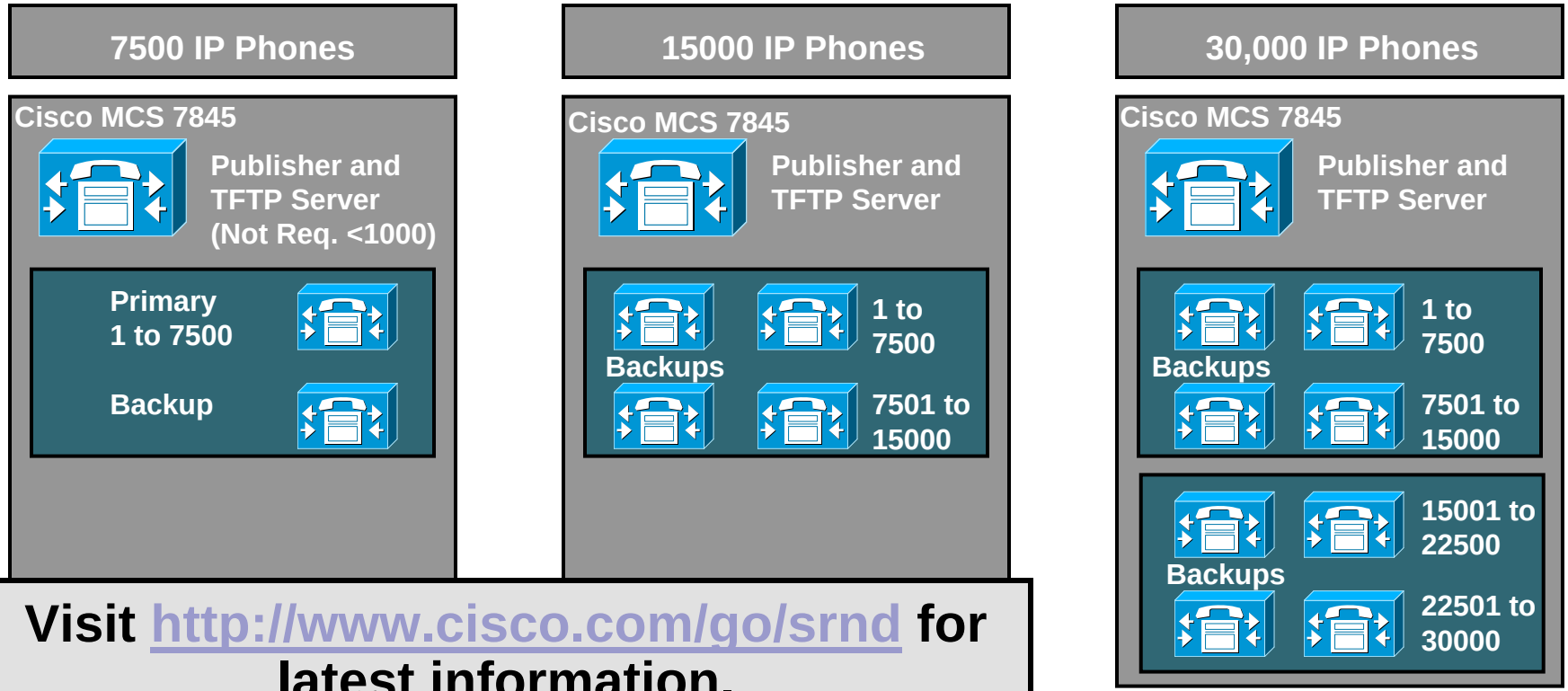
1:1 Redundancy Design



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



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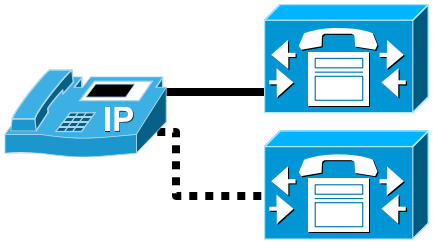
2:1 Redundancy Design



- 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.

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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

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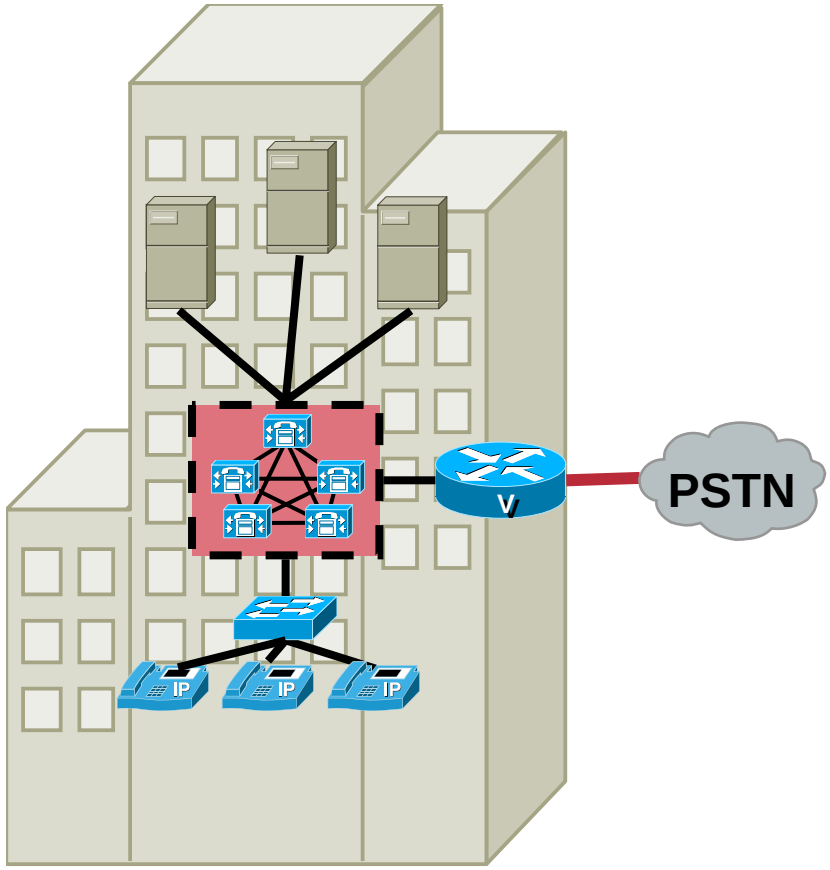
Single-Site Deployment

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**



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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**

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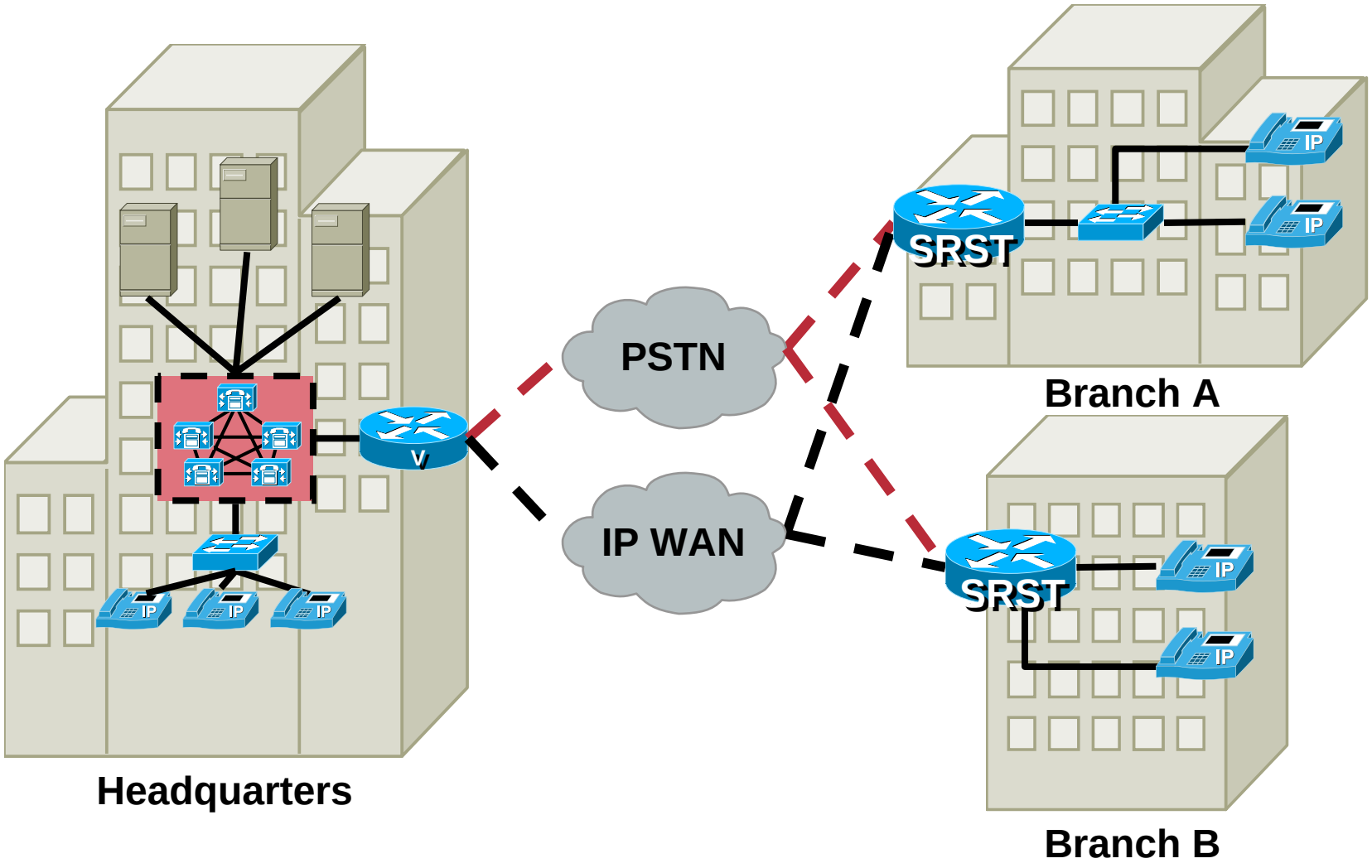
Multi site WAN with Centralized Call Processing



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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.**

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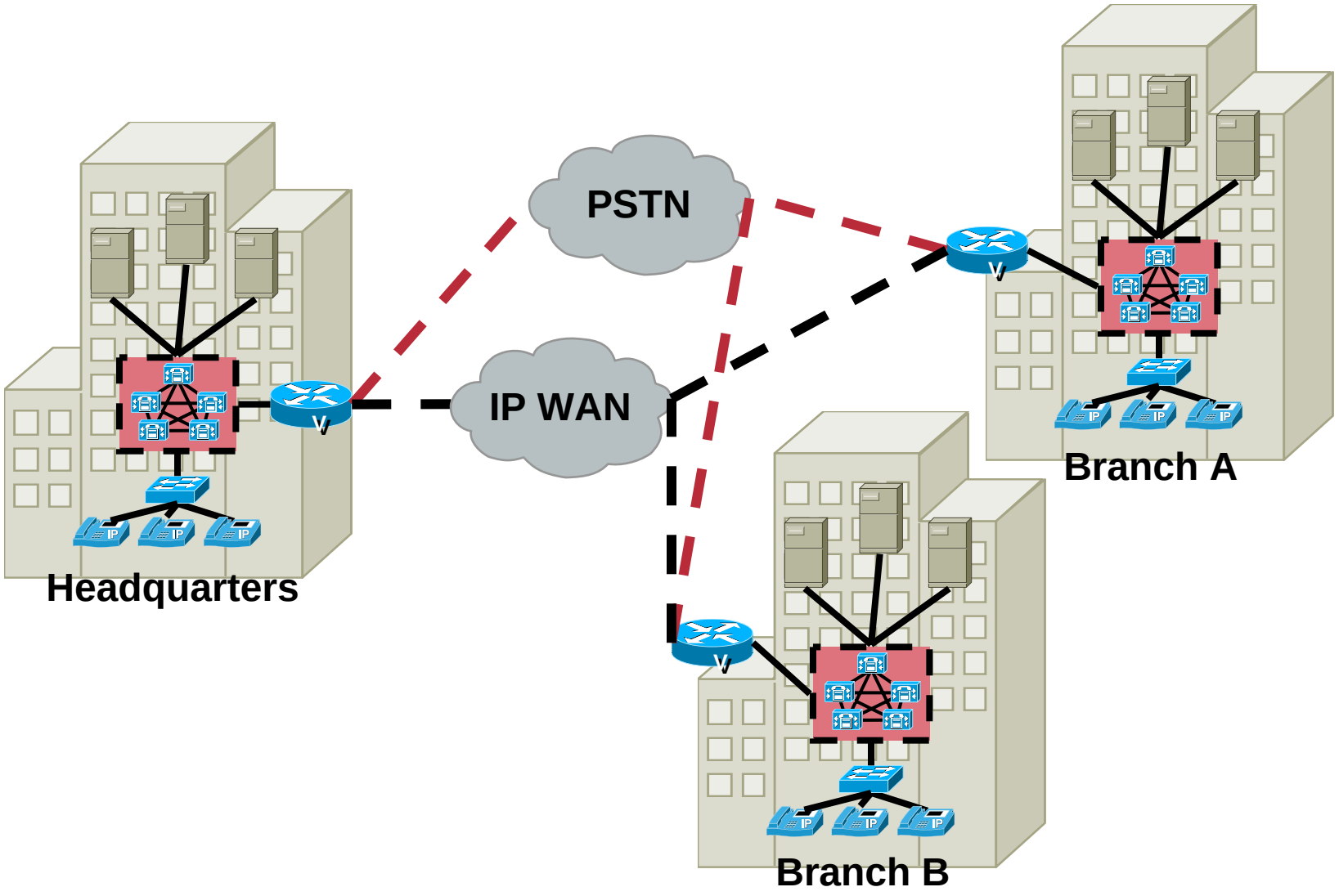
Multi site Distributed Call Processing: Topology



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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)**

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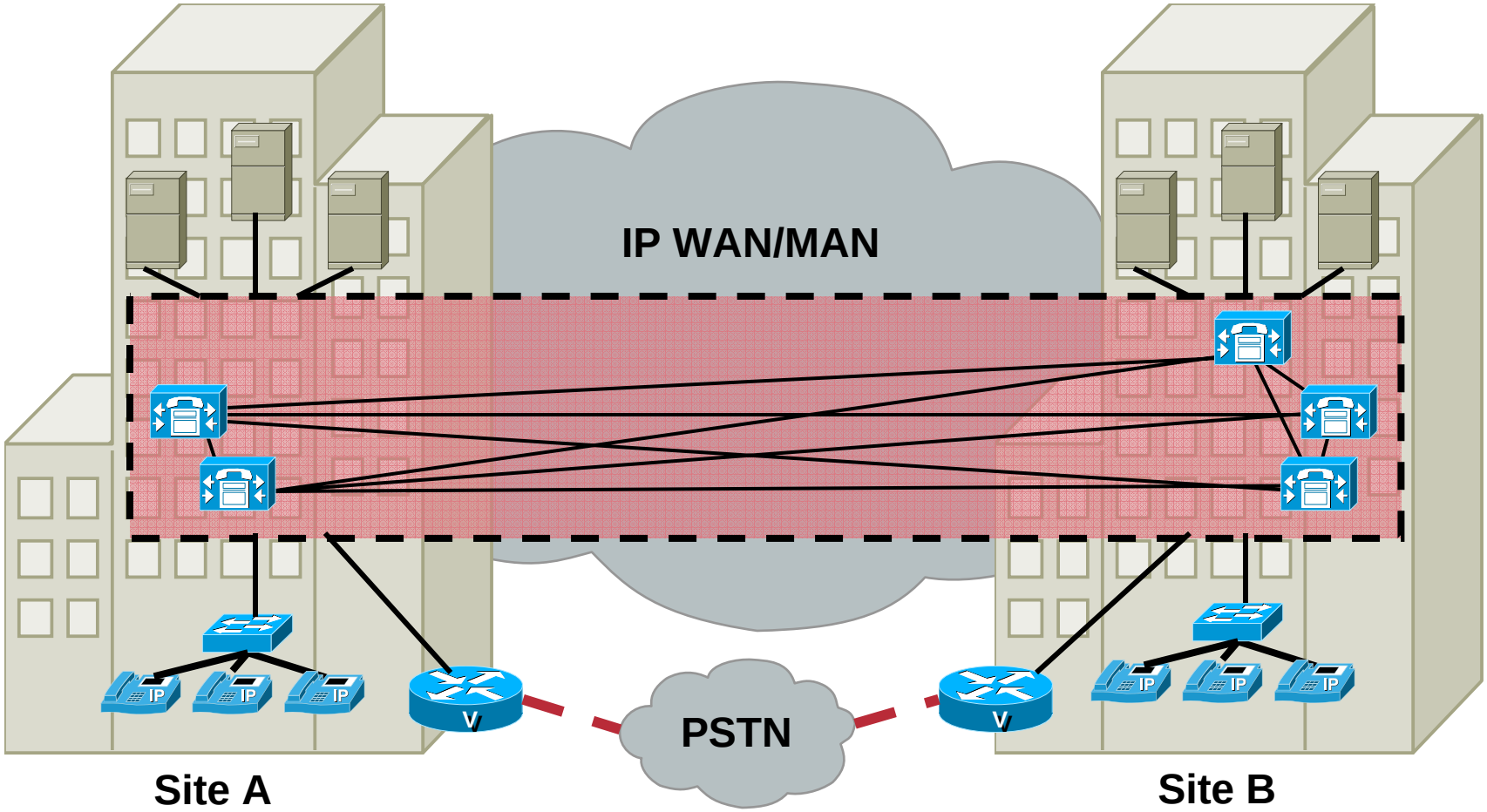
Clustering over the IP WAN/MAN: Topology



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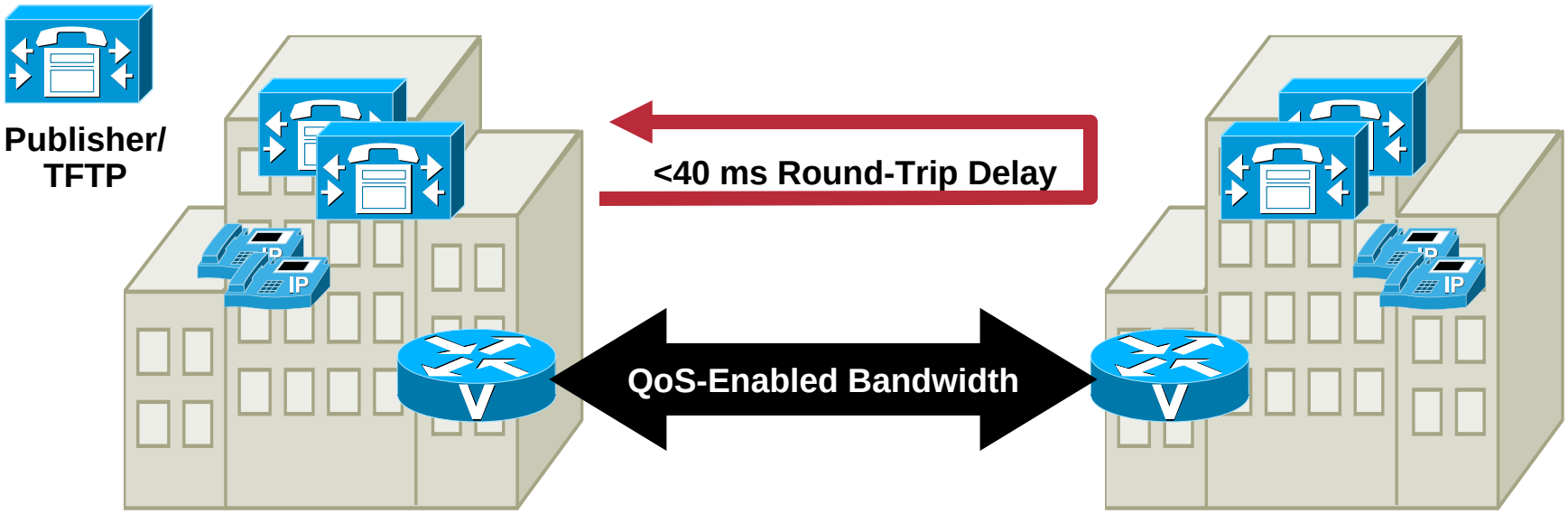
Clustering over the IP WAN: Requirements with CCM 4.x



- 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.

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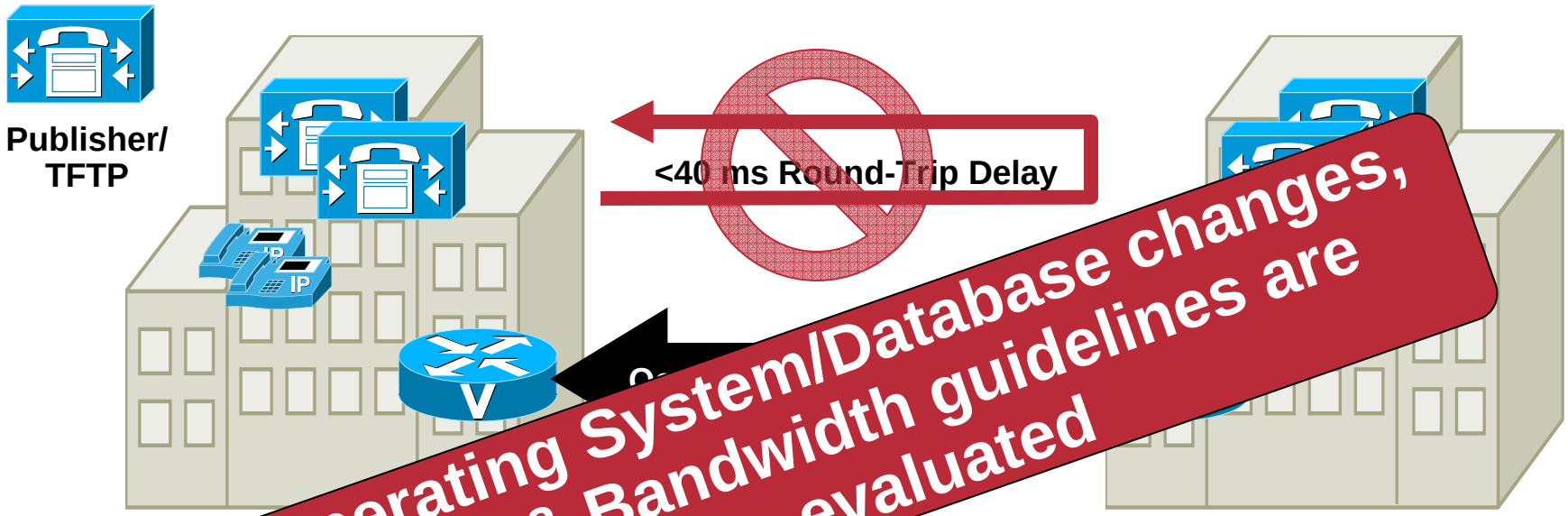
Clustering over the IP WAN: Requirements



Due Operating System/Database changes, RTT delay & Bandwidth guidelines are being re-evaluated

- **…lay between any two Cisco Unified C… servers in the cluster**
- **90… kbps for every 10,000 BHCAs within the cluster**

IP Communication SRND for CallManager 5.0 will have full details

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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.**

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Understanding Cisco Enterprise IP Communication

Evaluating Cisco IP Communication Endpoints

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Objectives

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

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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.

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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

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SCCP Phones

SCCP- only Phones

7902

7920

7936

7985

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SCCP Devices

SCCP- only Devices

Communicator **ATA-18x** **VG248/224**

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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.

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SIP Endpoints (Basic)

“Existing” Basic Cisco SIP Phones

Very Basic SIP

7912

7940/7960

Generic 3rd-Party

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SIP Endpoints (Advanced)

“Enhanced” SIP Phones

PC-Application

7911 **7941/7961** **7970/7971** **Cisco Unified Personal Communicator**

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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.**

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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**

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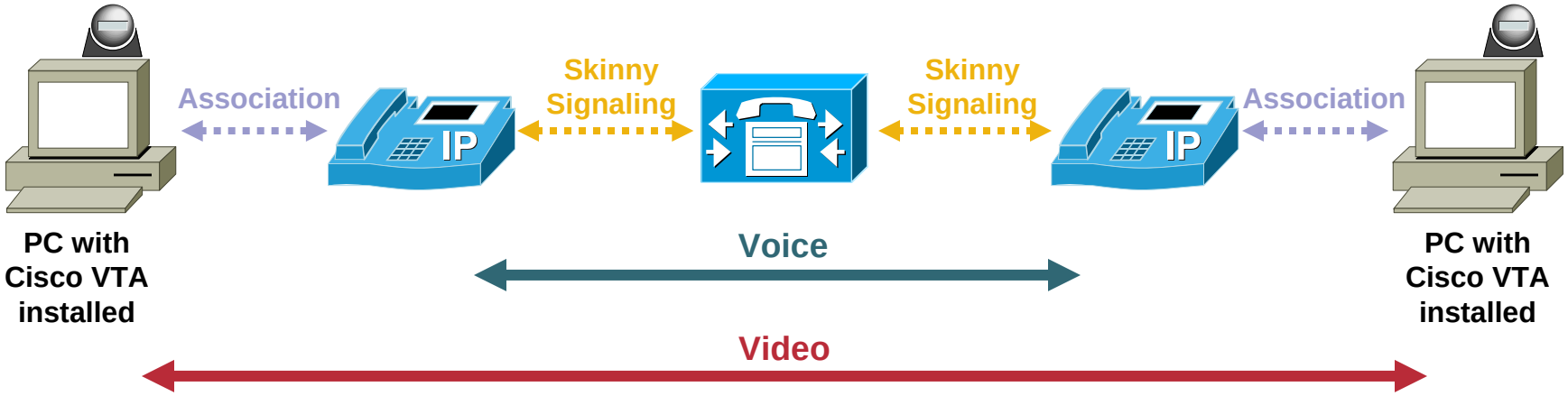
Cisco VTA Component Interaction



- 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

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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.

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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**

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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**

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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.**

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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

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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.

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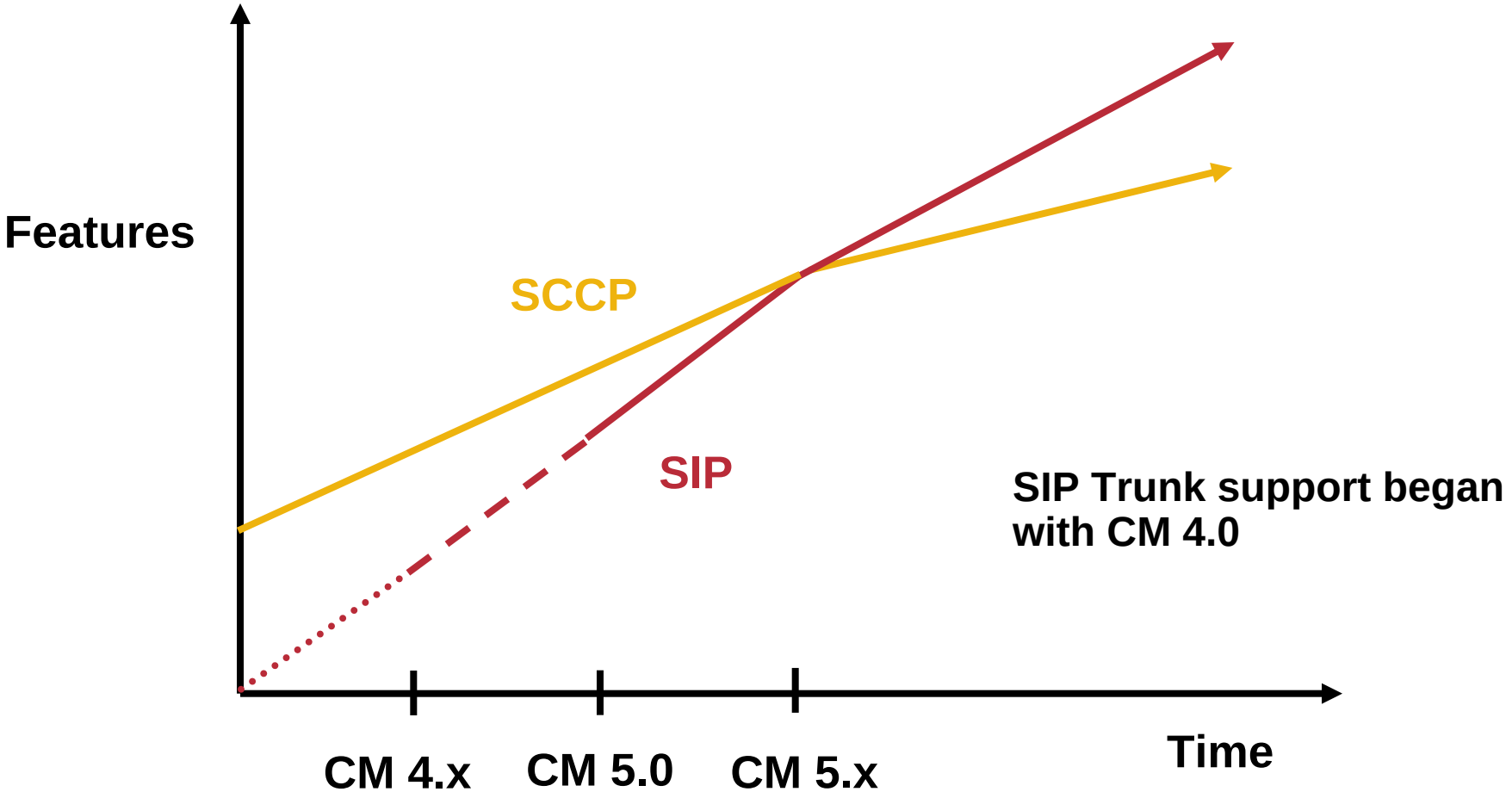
Feature Richness SCCP - SIP



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SIP vs SCCP configuration differences

SIP	SCCP
• Configuration file via TFTP • Softkey File via TFTP (Enhanced IP Phones only)	• Configuration via TFTP • 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

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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

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Deploying a Cisco Unified CallManager 5.0 Cluster

Cisco Unified CallManager 5.0 Architecture

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Objectives

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

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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).

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Cisco Unified CallManager 5.0 Overview

Single Installation Media

RedHat Linux as OS

Cisco Unified CallManager 5.0 as application

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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.**

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Appliance Model

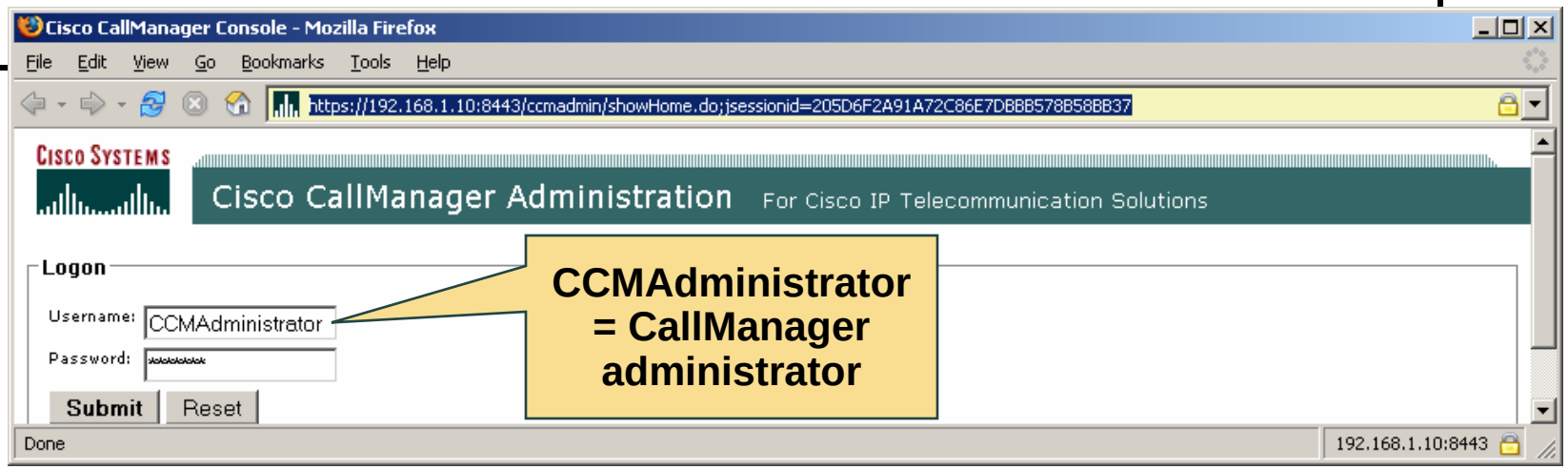
```
login as: administrator
administrator@192.168.1.10's password:

   Welcome to the Platform Command Line Interface (version 1.1)

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

admin:
```

Administrator = Platform administrator



CCMAdministrator = CallManager administrator

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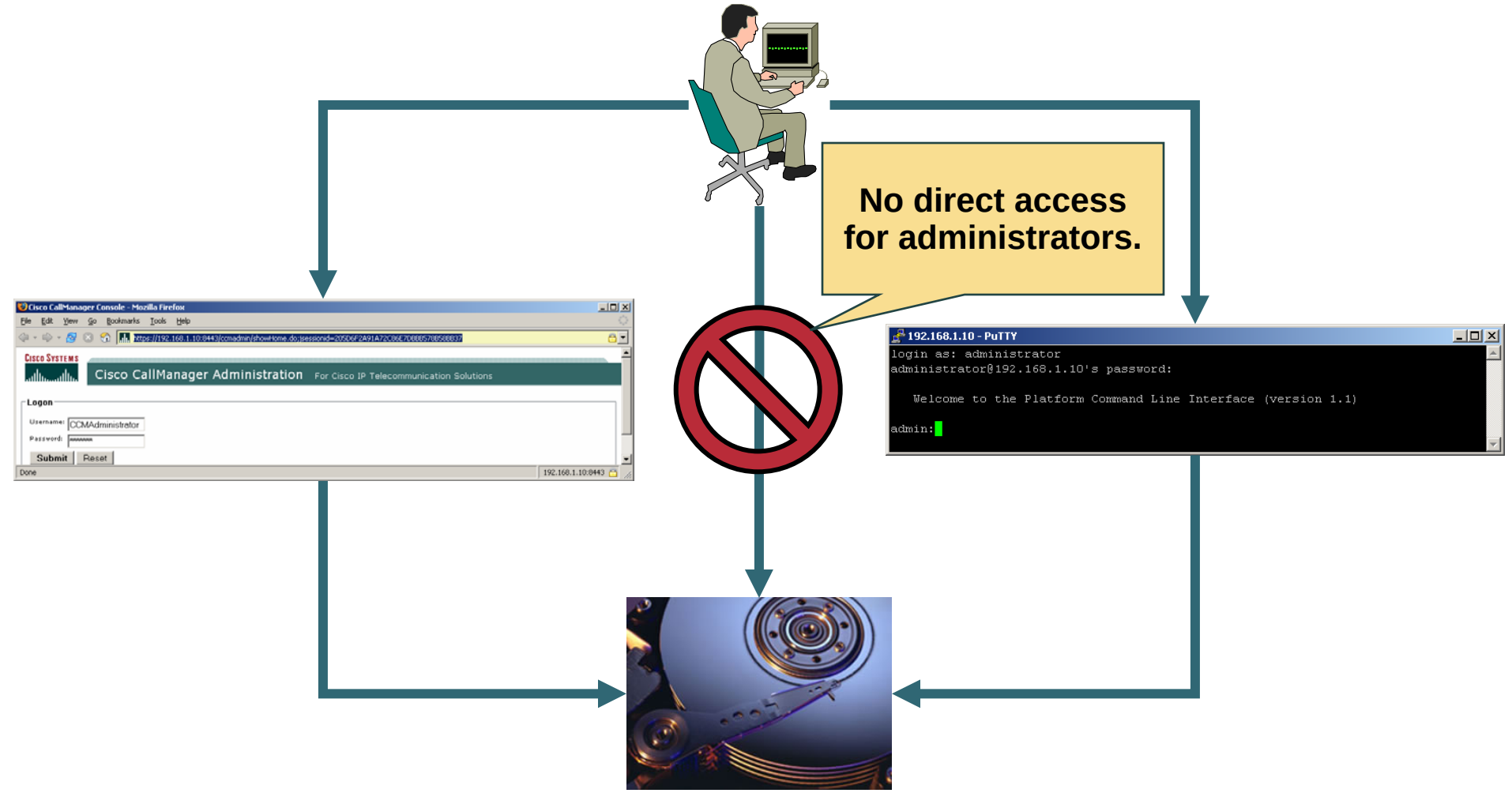
Appliance Model



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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.**

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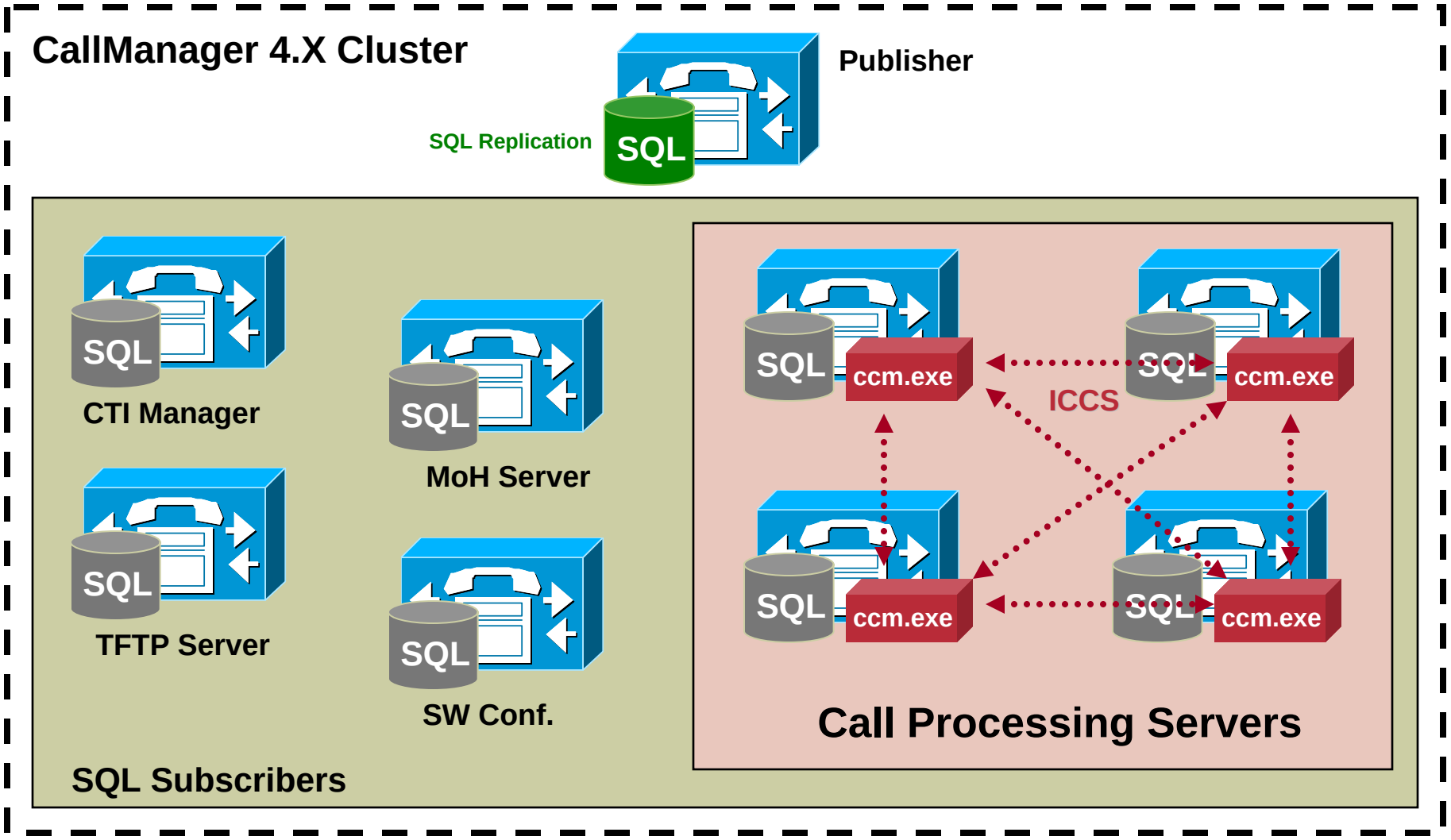
Database Changes



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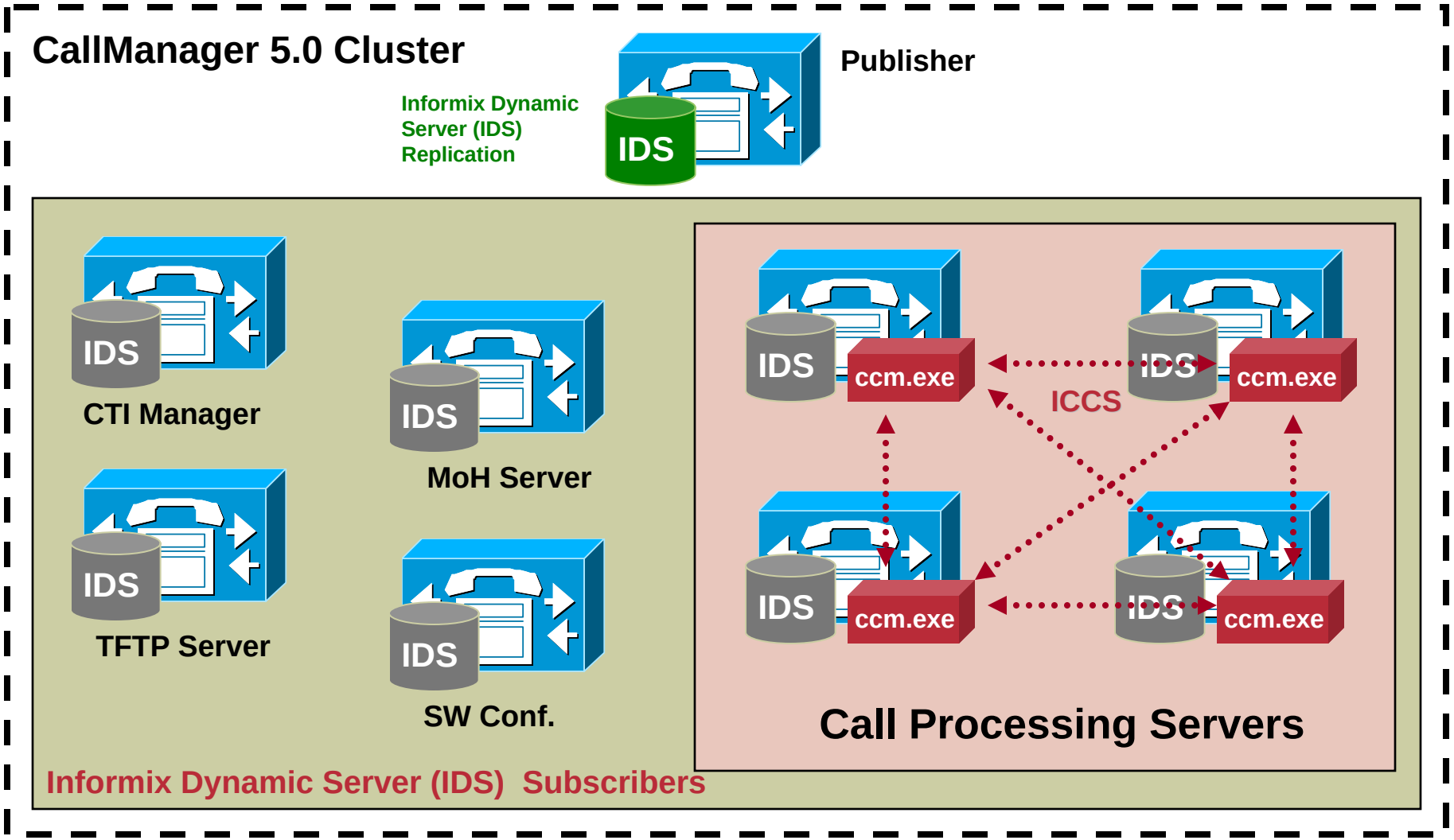
Database Changes



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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**

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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*
```

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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.

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Summary

- Cisco Unified CallManager 5.0 is based on a Linux appliance.
- The appliance seperates appliaction 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.

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Deploying a Cisco Unified CallManager 5.0 Cluster

Installing Cisco Unified CallManager 5.0

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Objectives

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

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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.**

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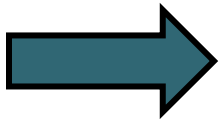
Hardware Requirements (cont.)

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.

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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

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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 CCMAdministrator

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Basic Installation Procedure

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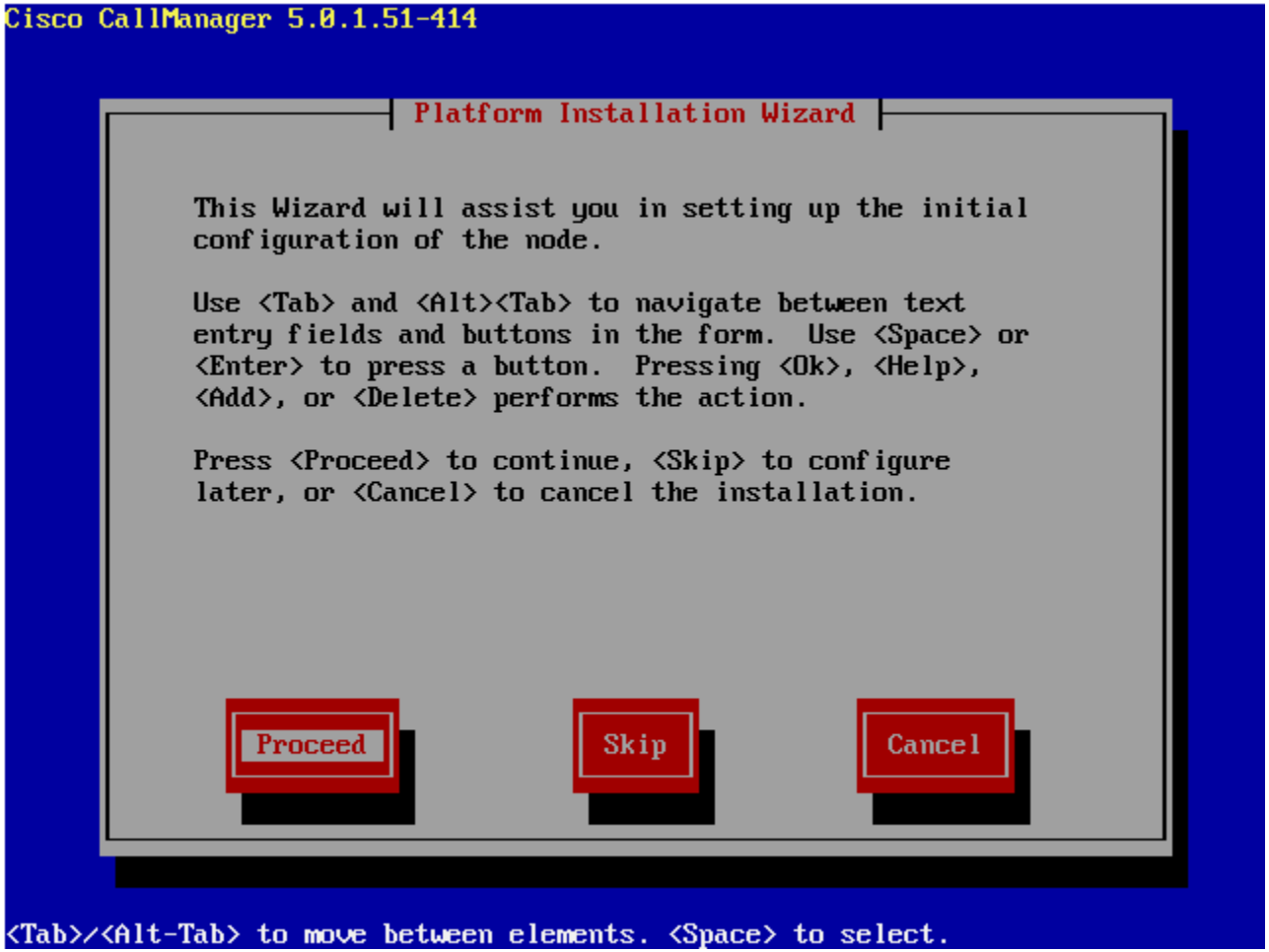
Basic Installation Procedure



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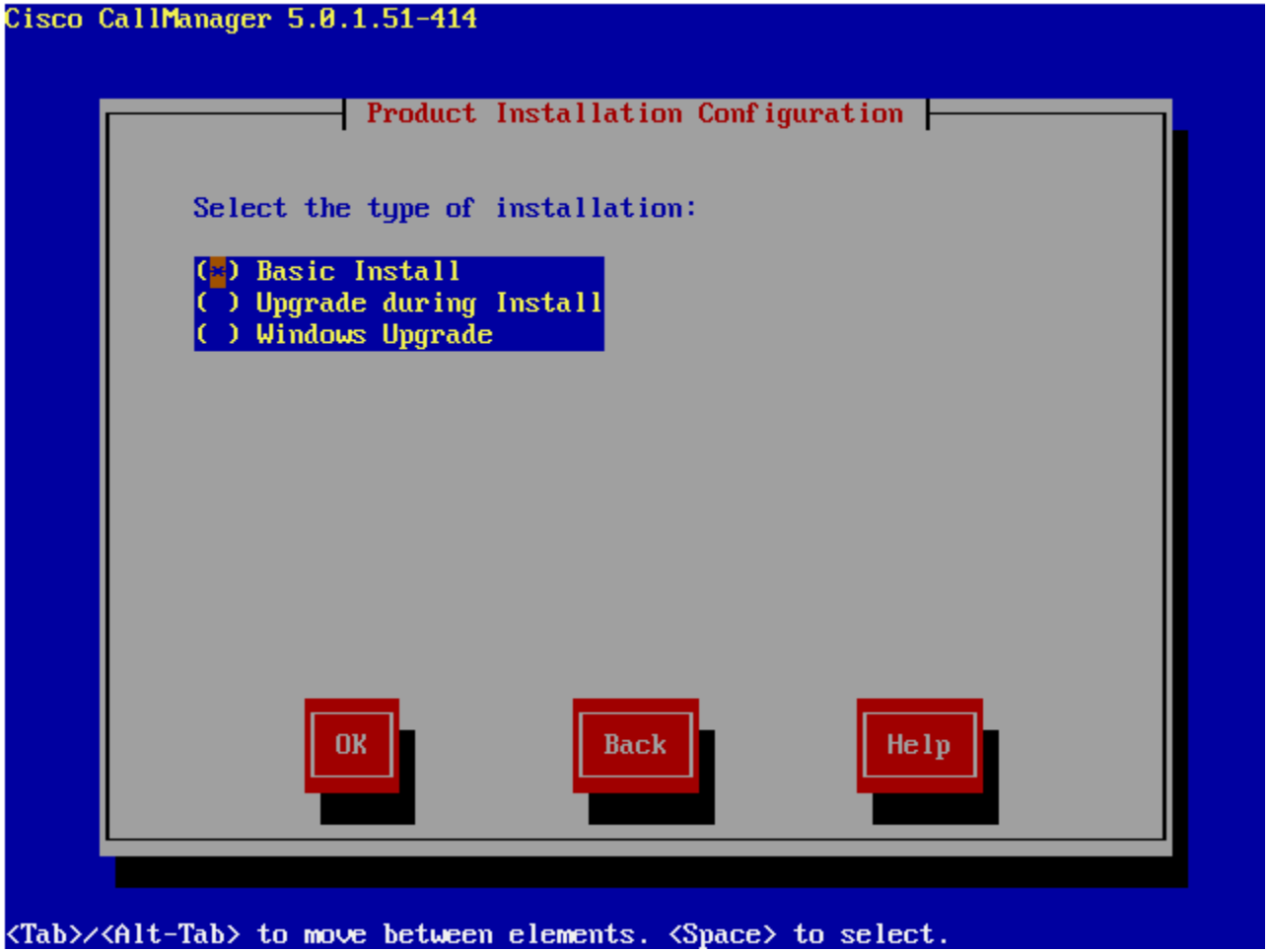
Basic Installation Procedure



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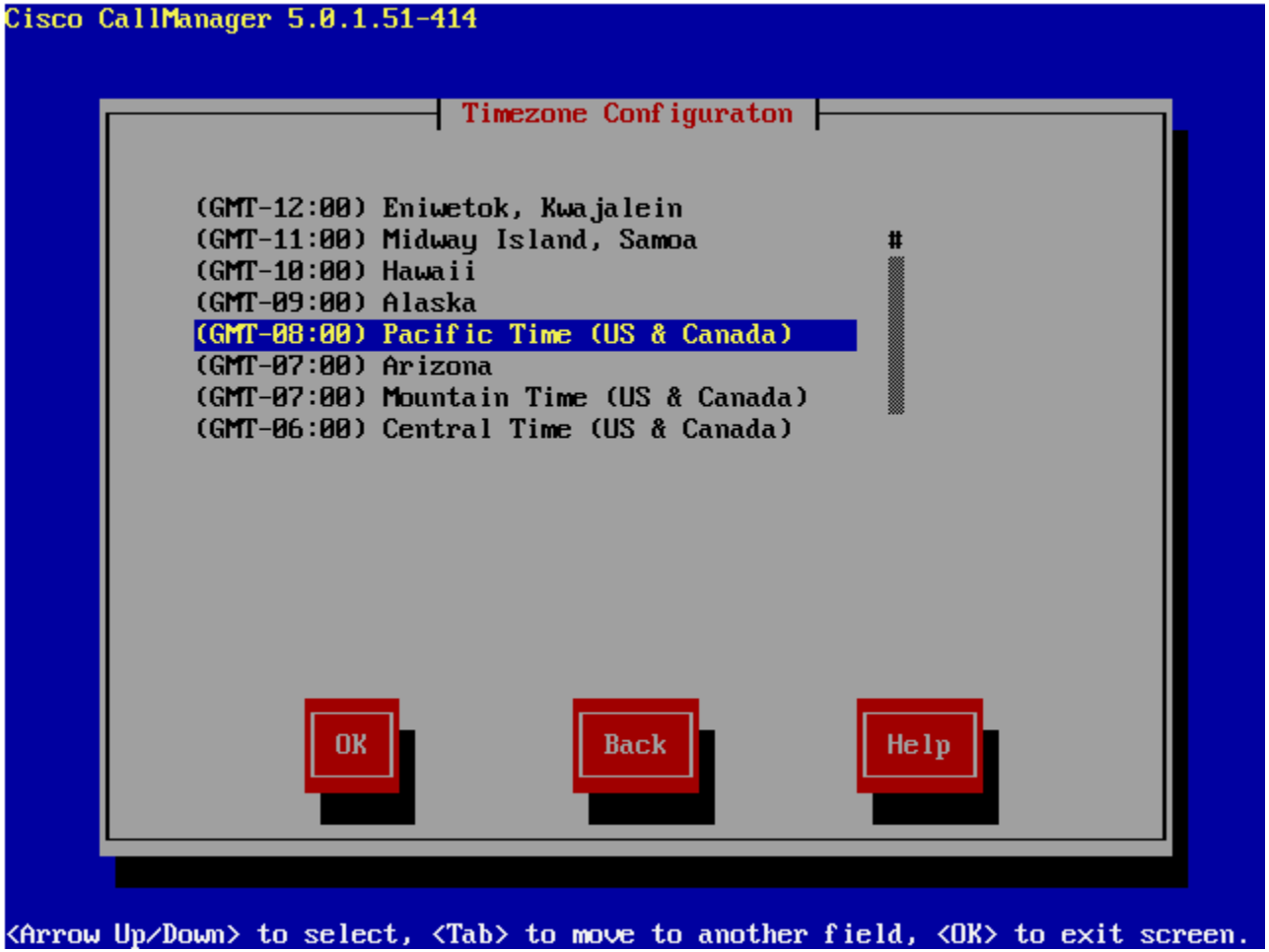
Timezone Configuration



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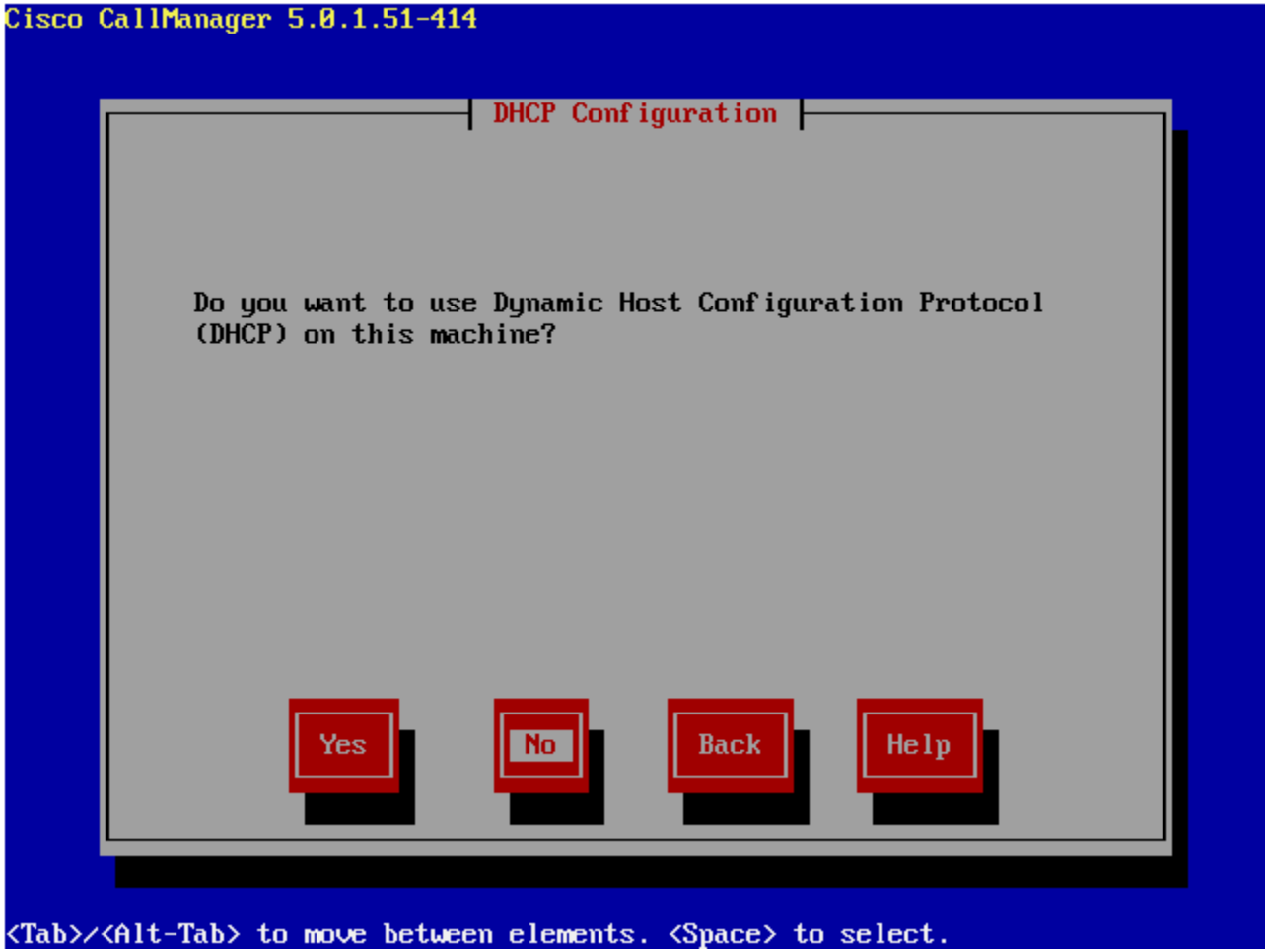
Static Network Configuration



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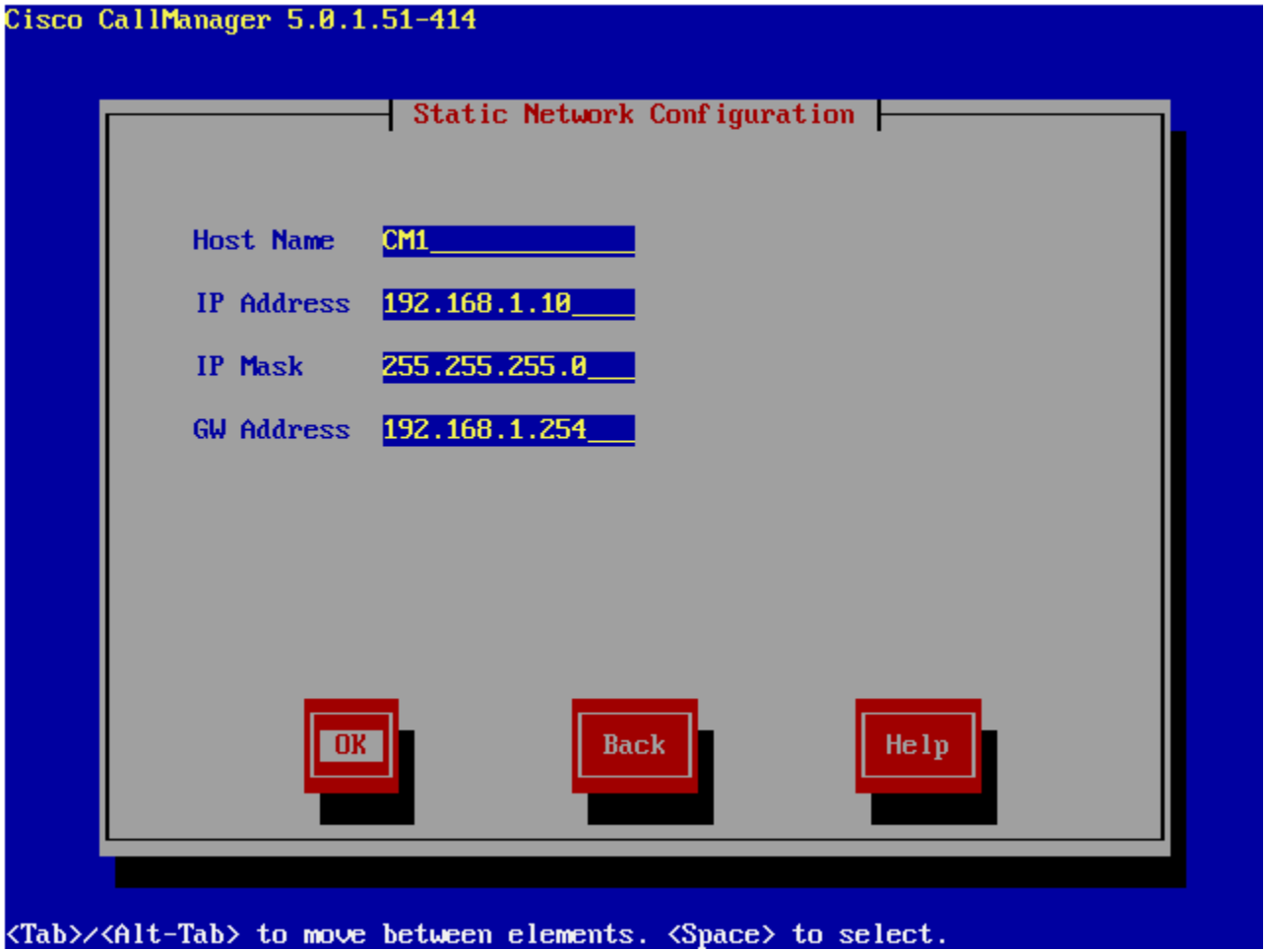
Static Network Configuration



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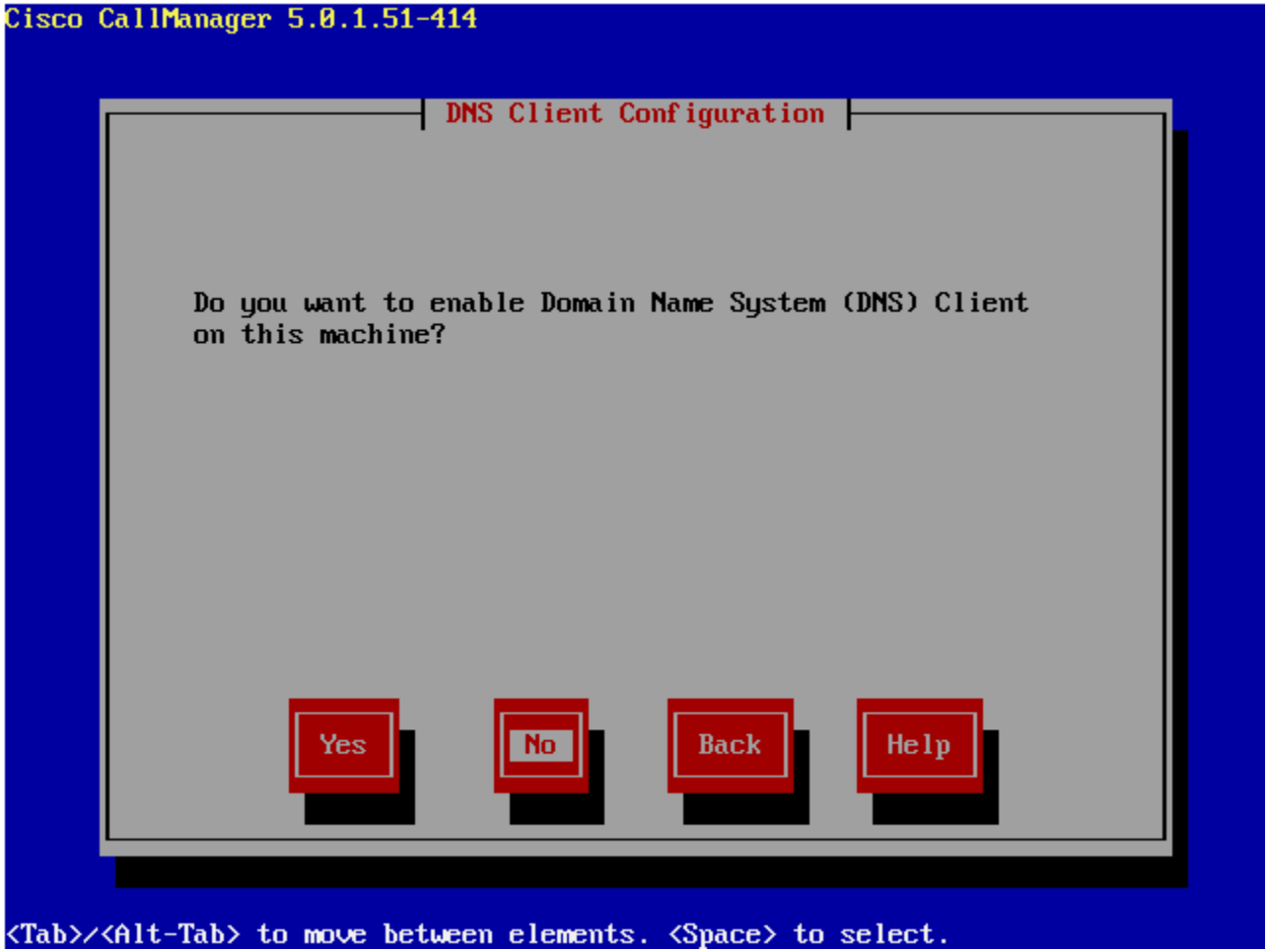
DNS Client Configuration



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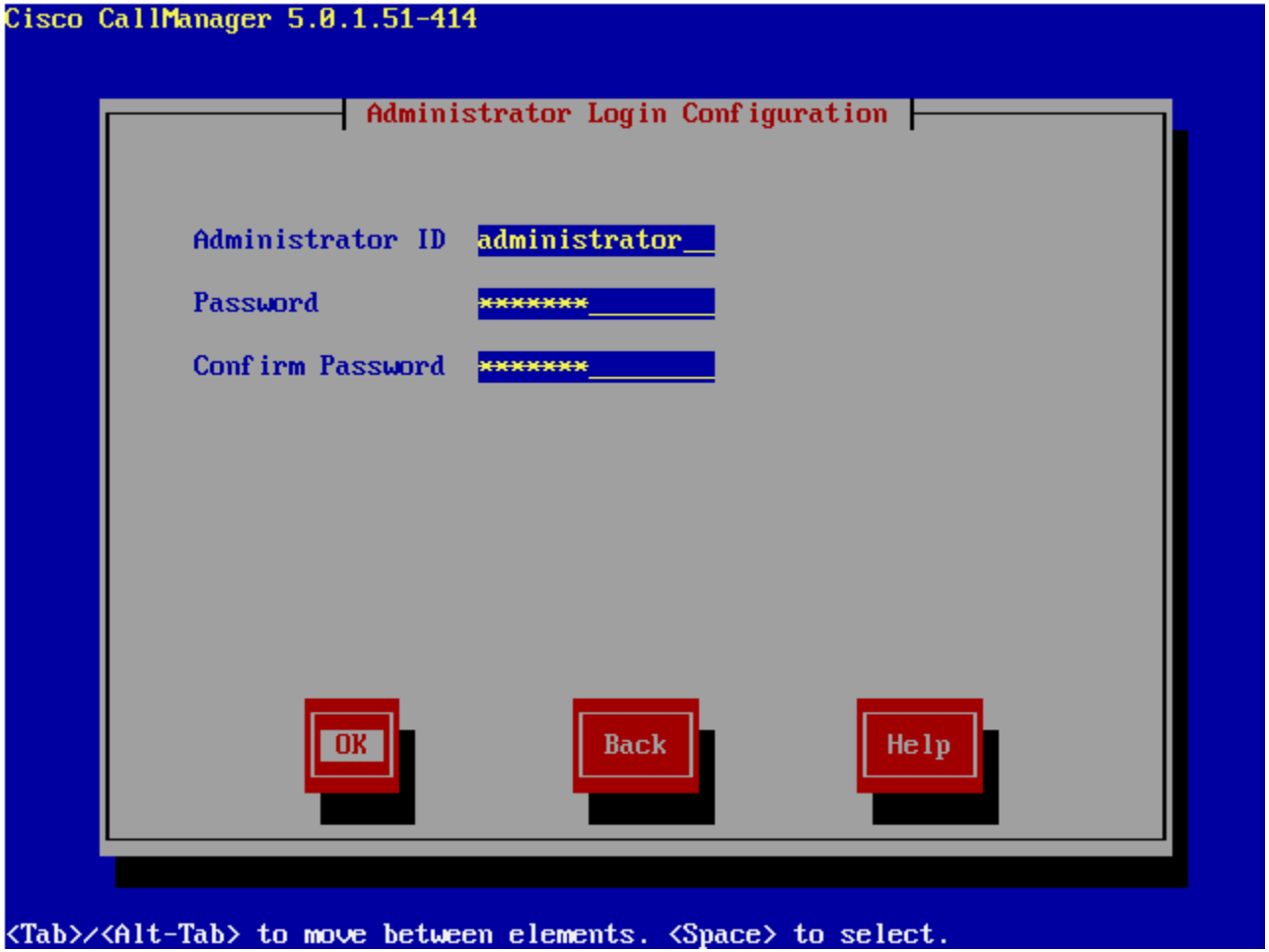
Administrator Login Configuration



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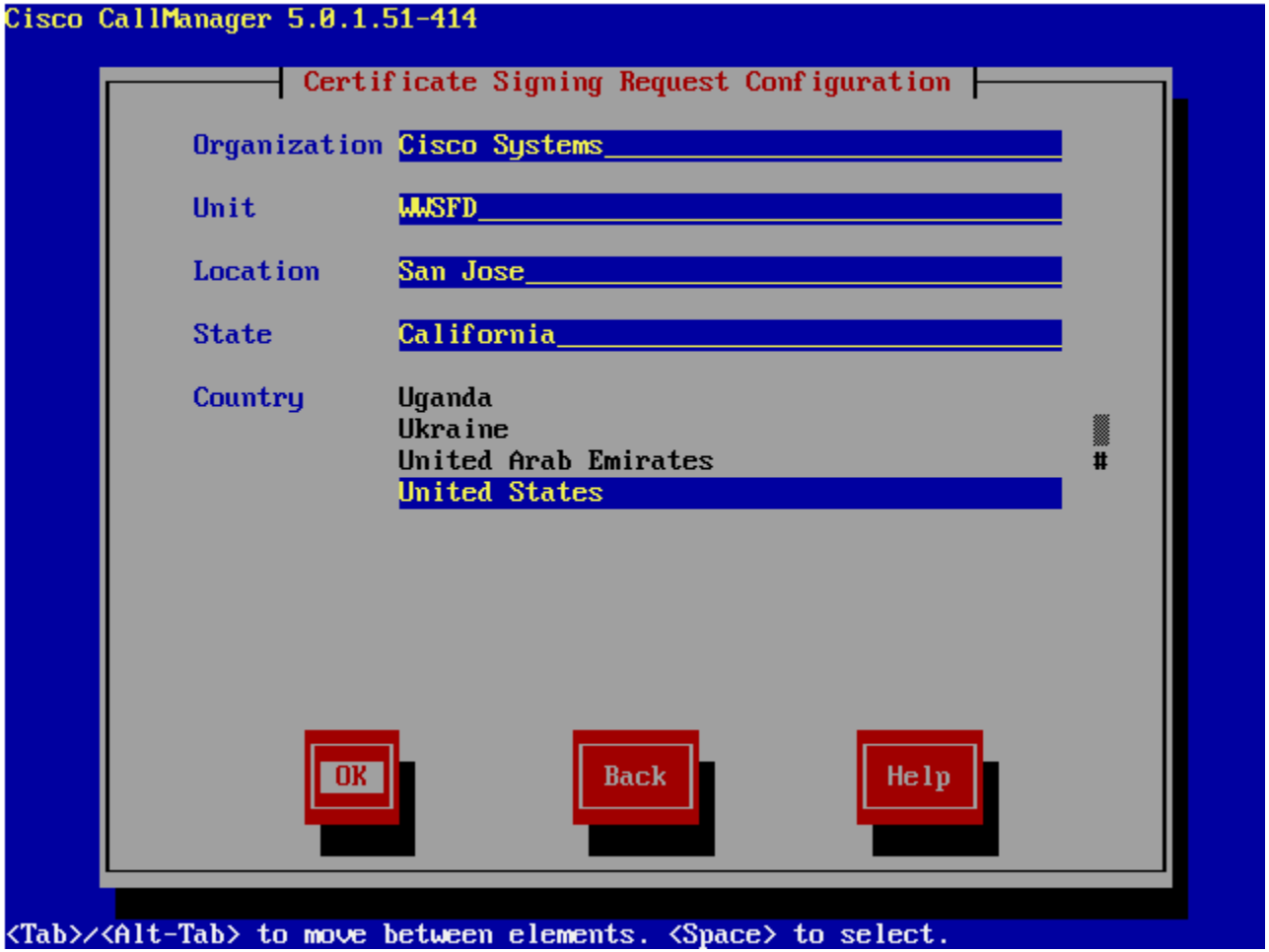
Certificate Configuration



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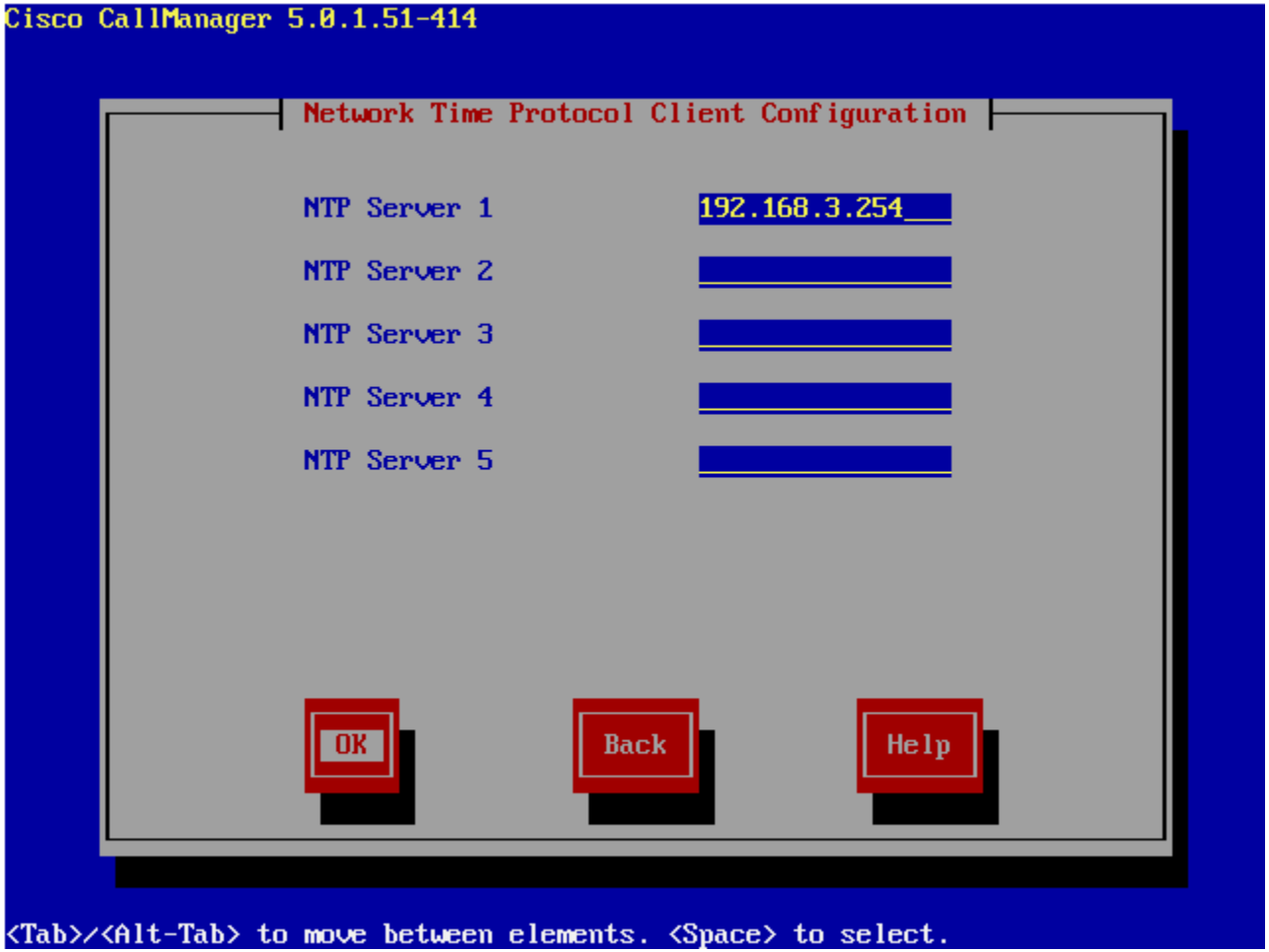
NTP Client Configuration



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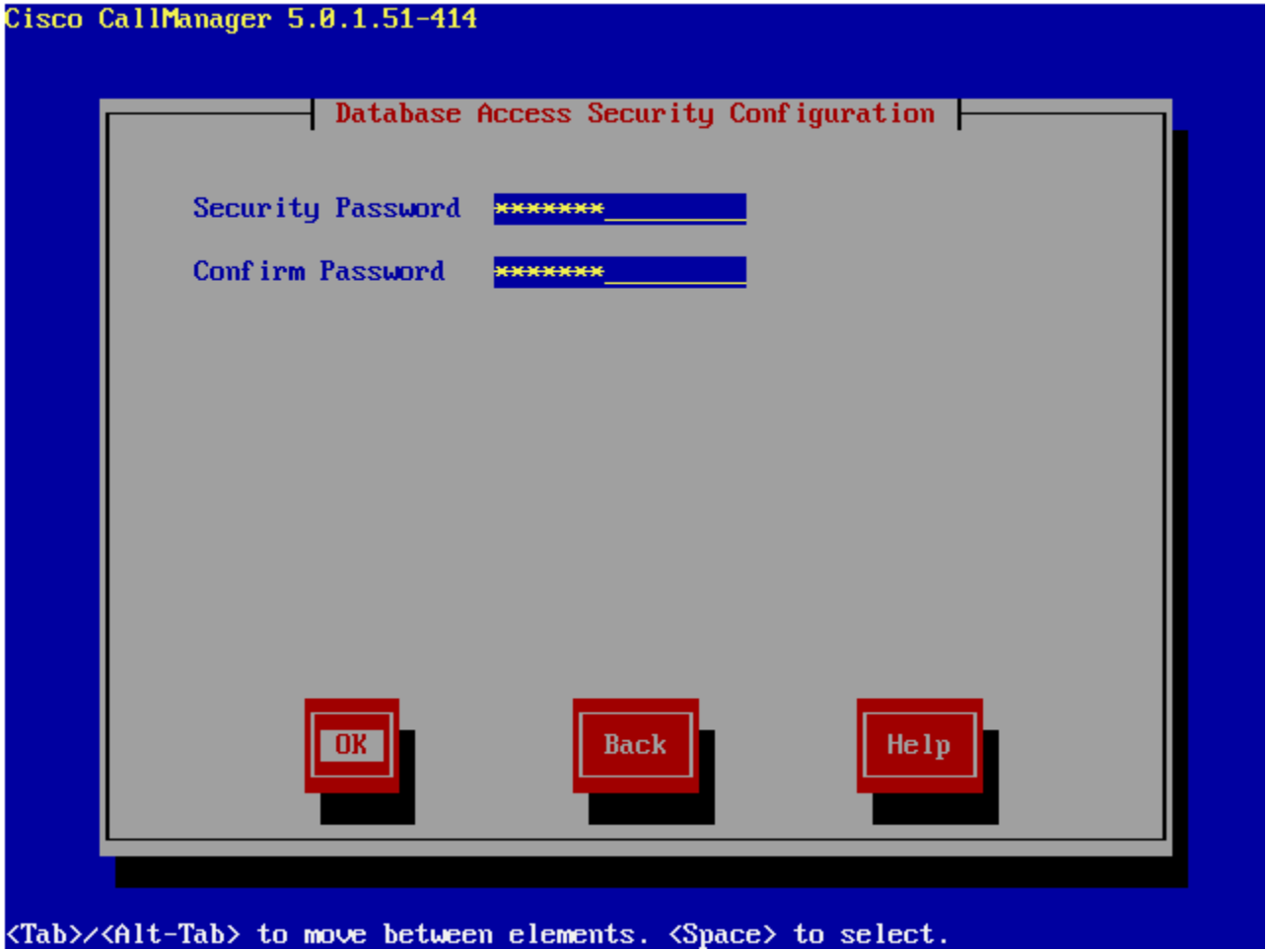
Database Access Security Configuration



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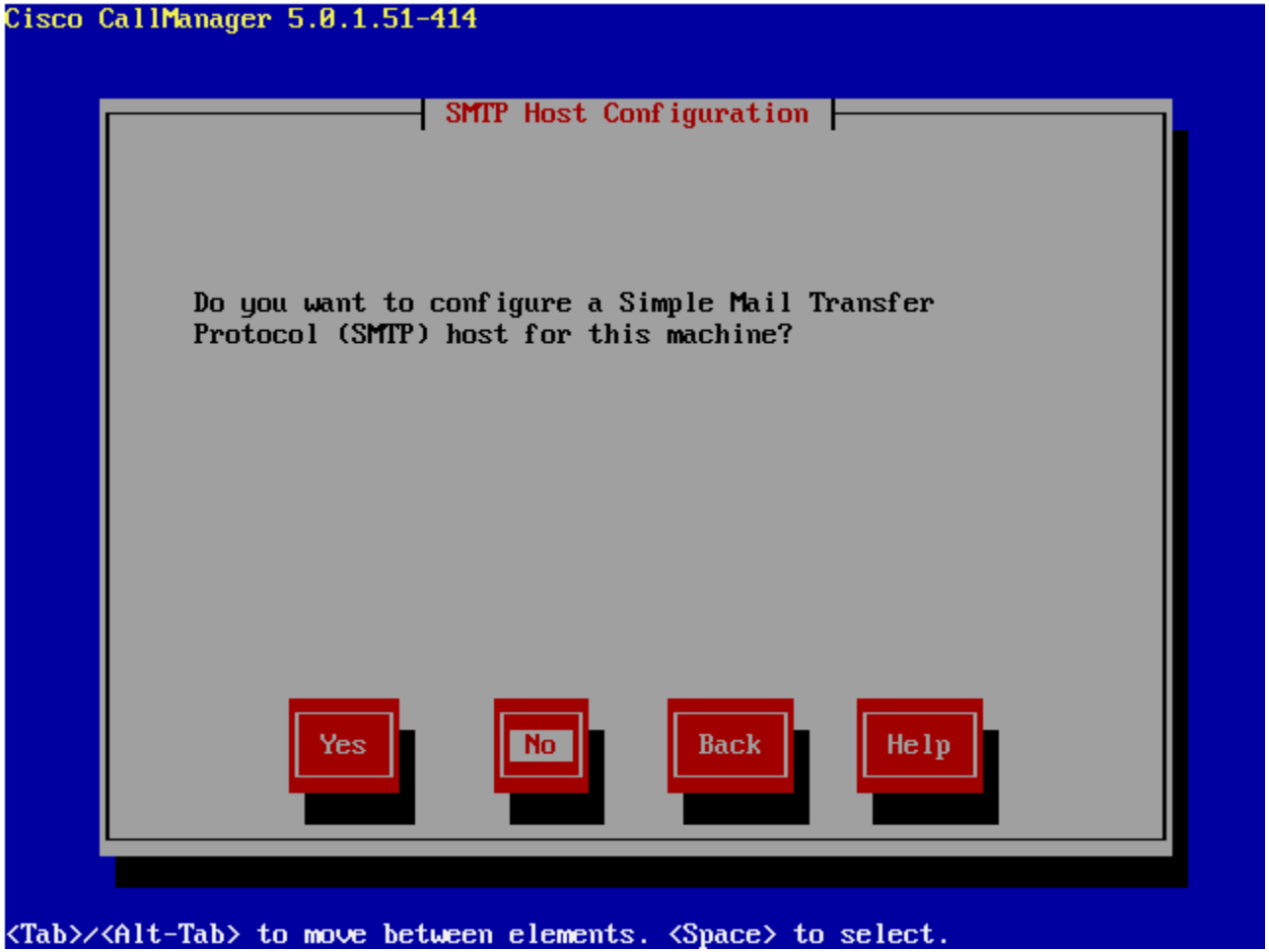
SMTP Configuration



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CCMAdministrator Password Configuration

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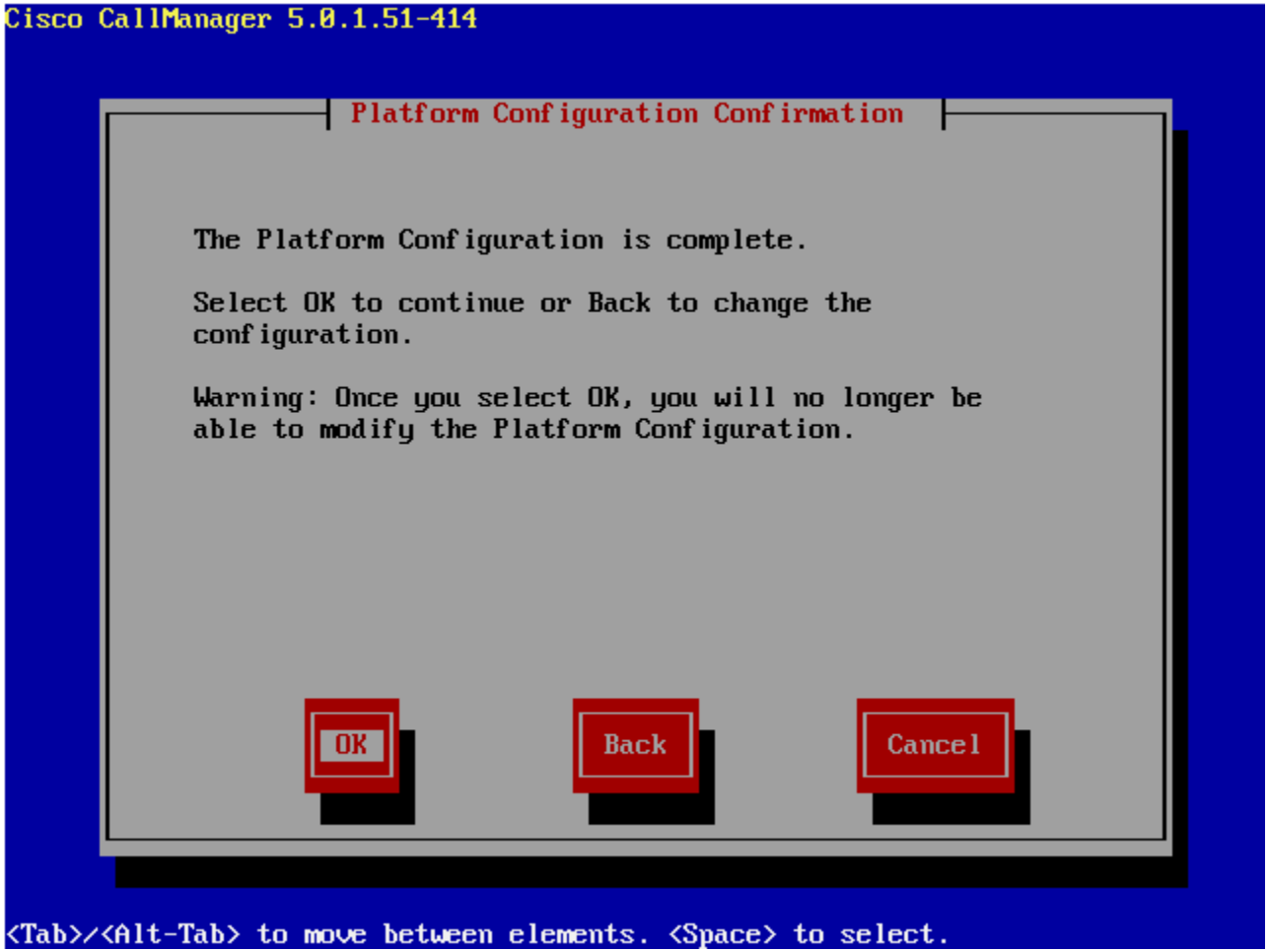
Confirm Platform Configuration



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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.

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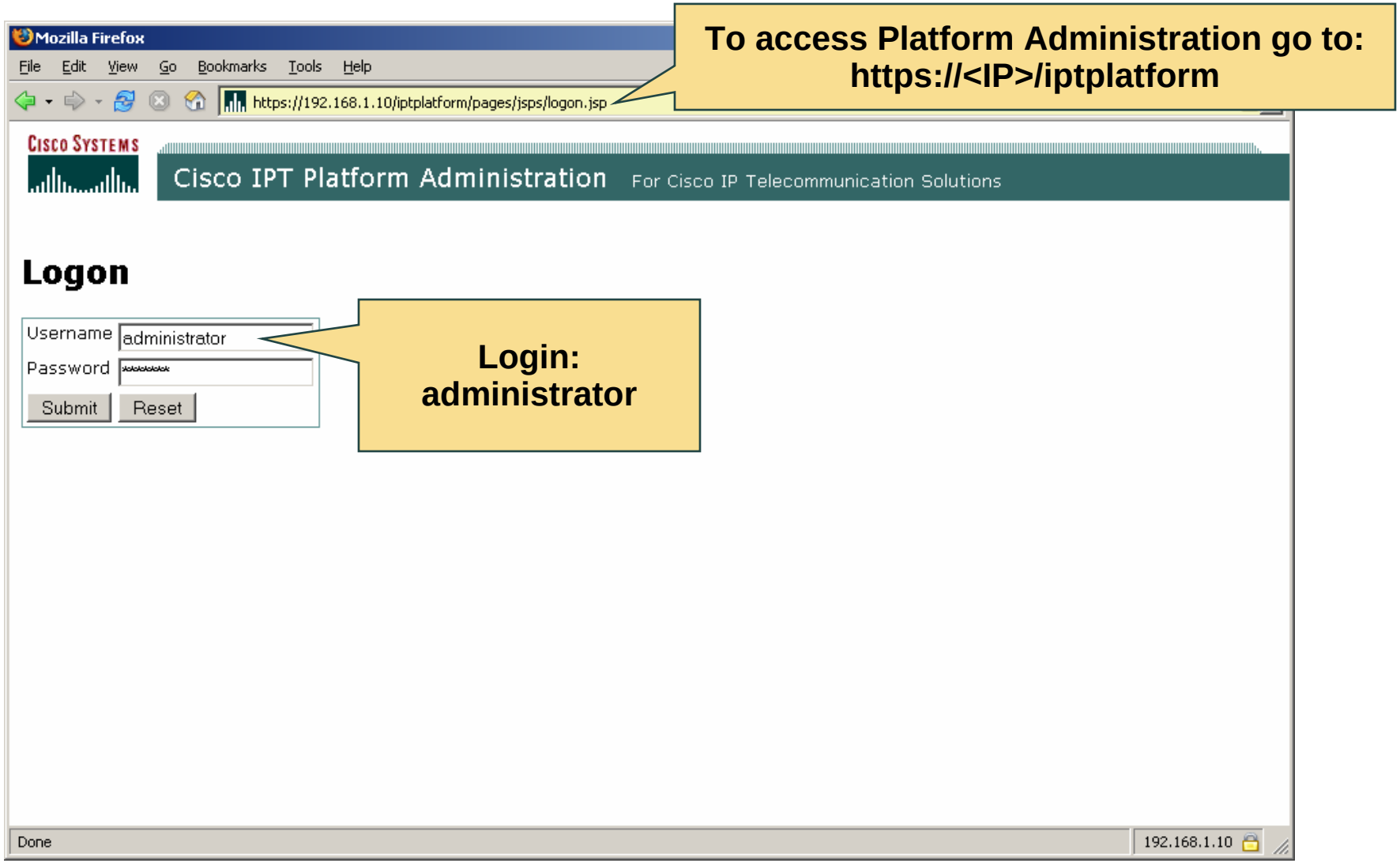
Slide 21

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

Obtain License File: Publisher MAC Address



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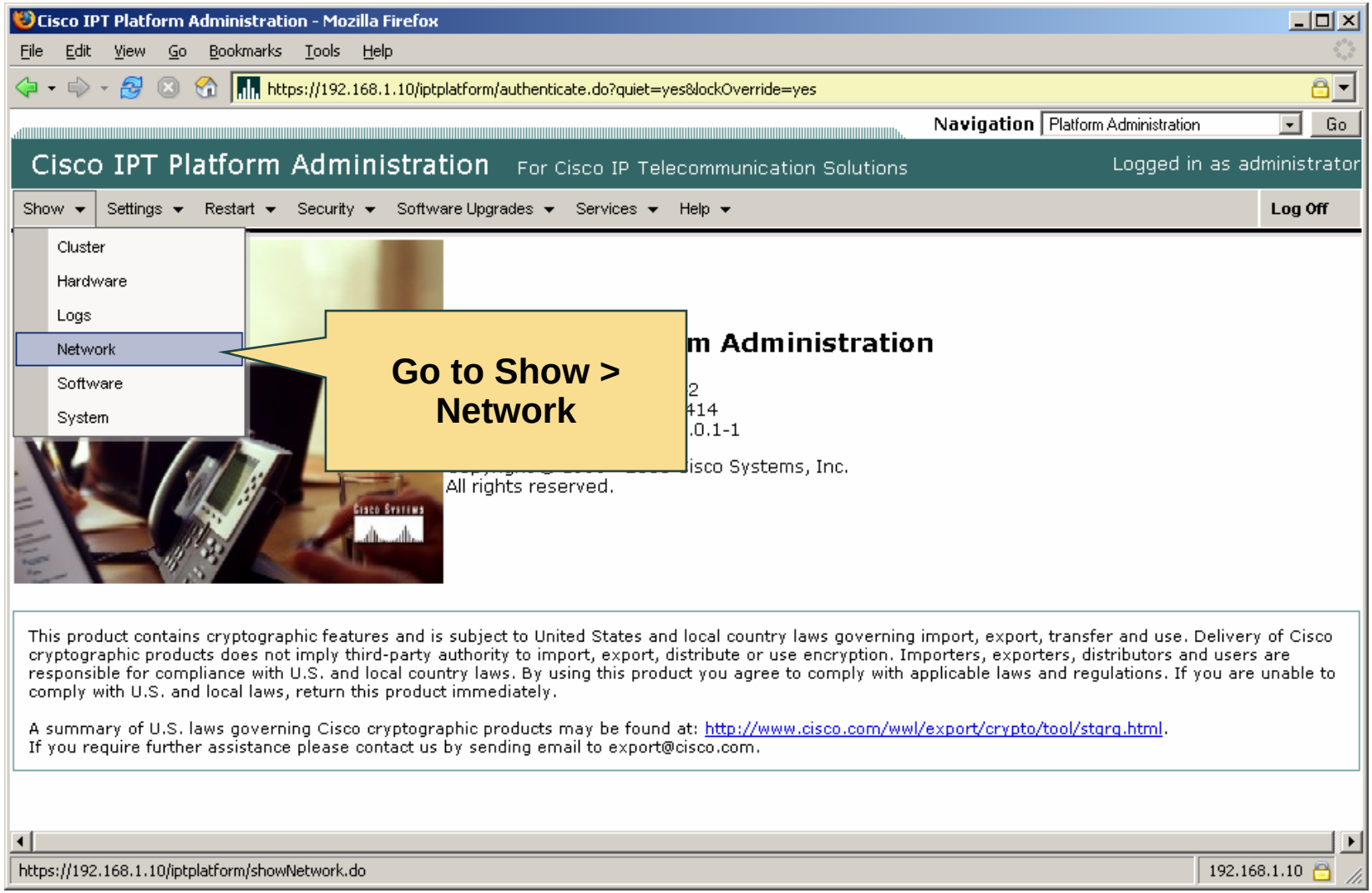
Obtain License File: Publisher MAC Address



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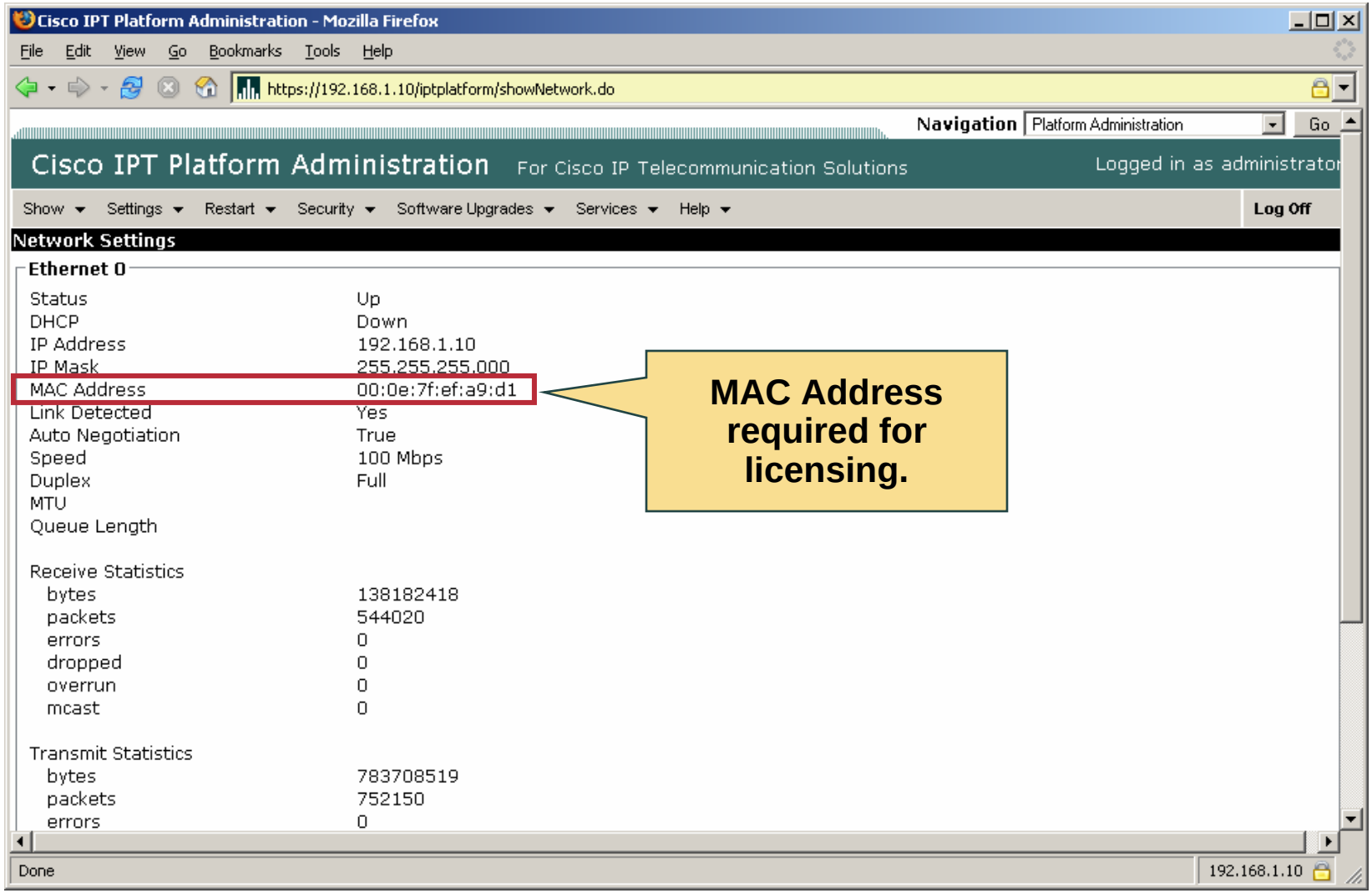
Obtain License File: Publisher MAC Address



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Install License File

Installing License Files:

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

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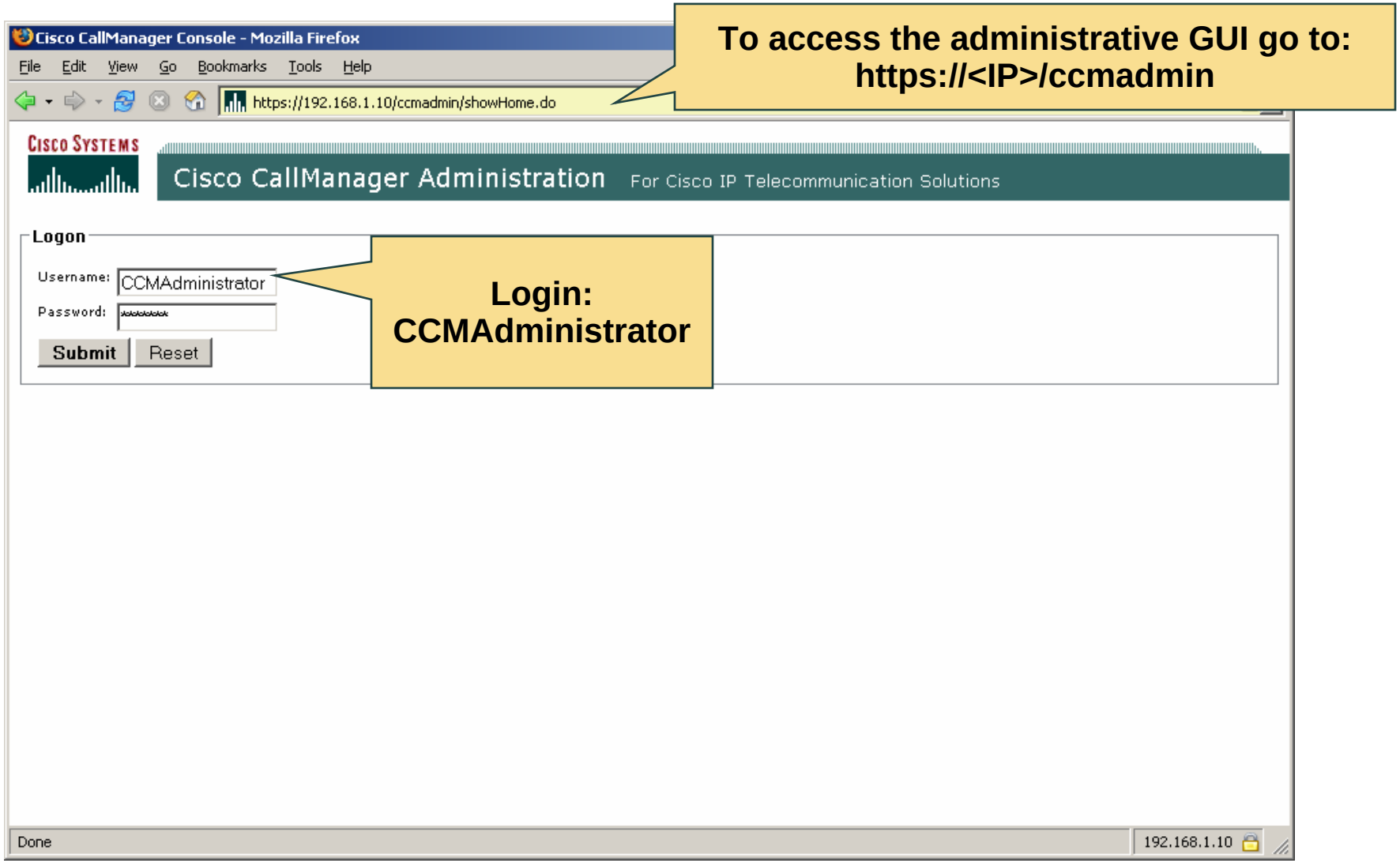
Install License File: Access CCMAdmin Webpage



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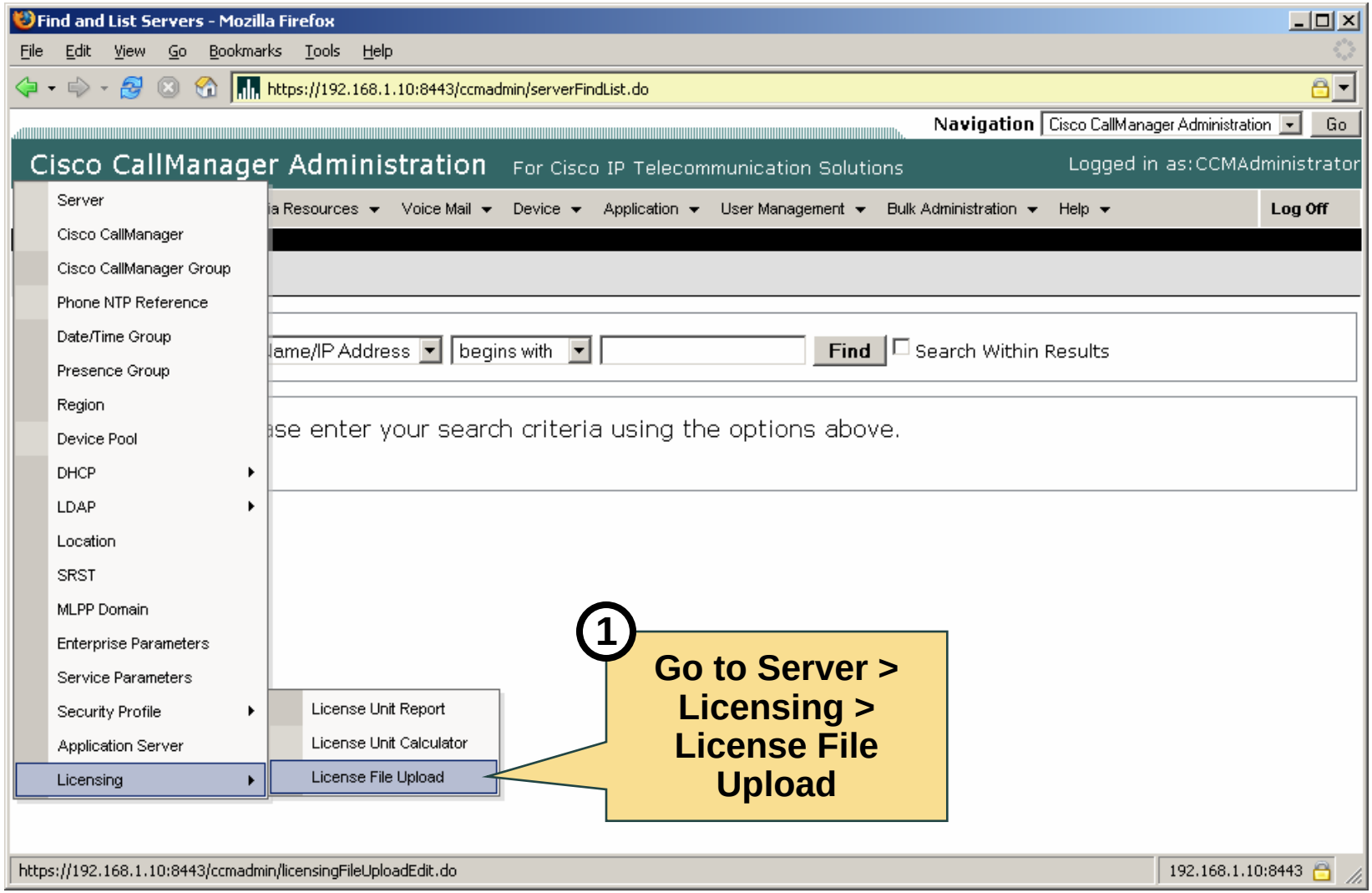
Install License File: Upload License File



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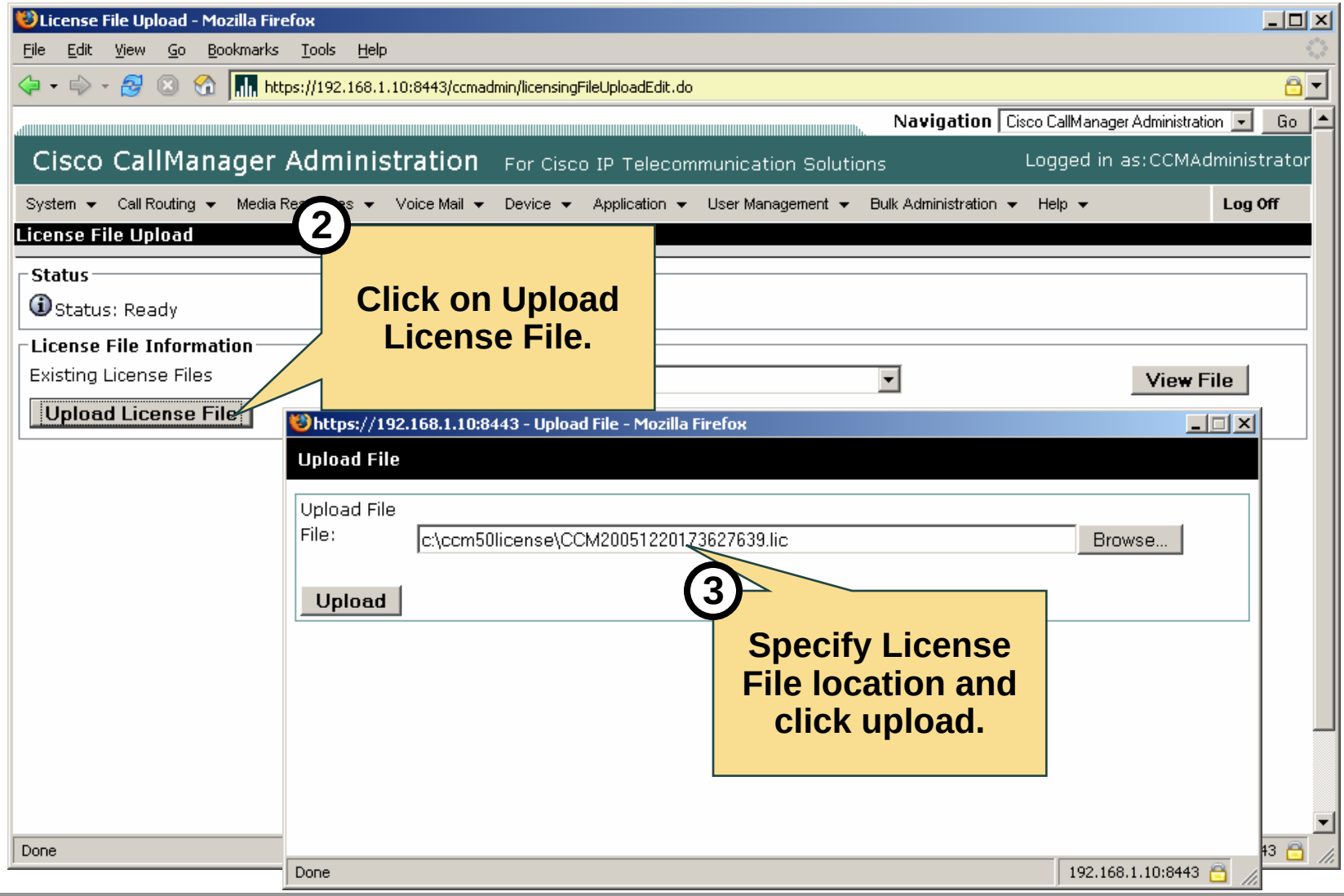
Install License File: Upload License File



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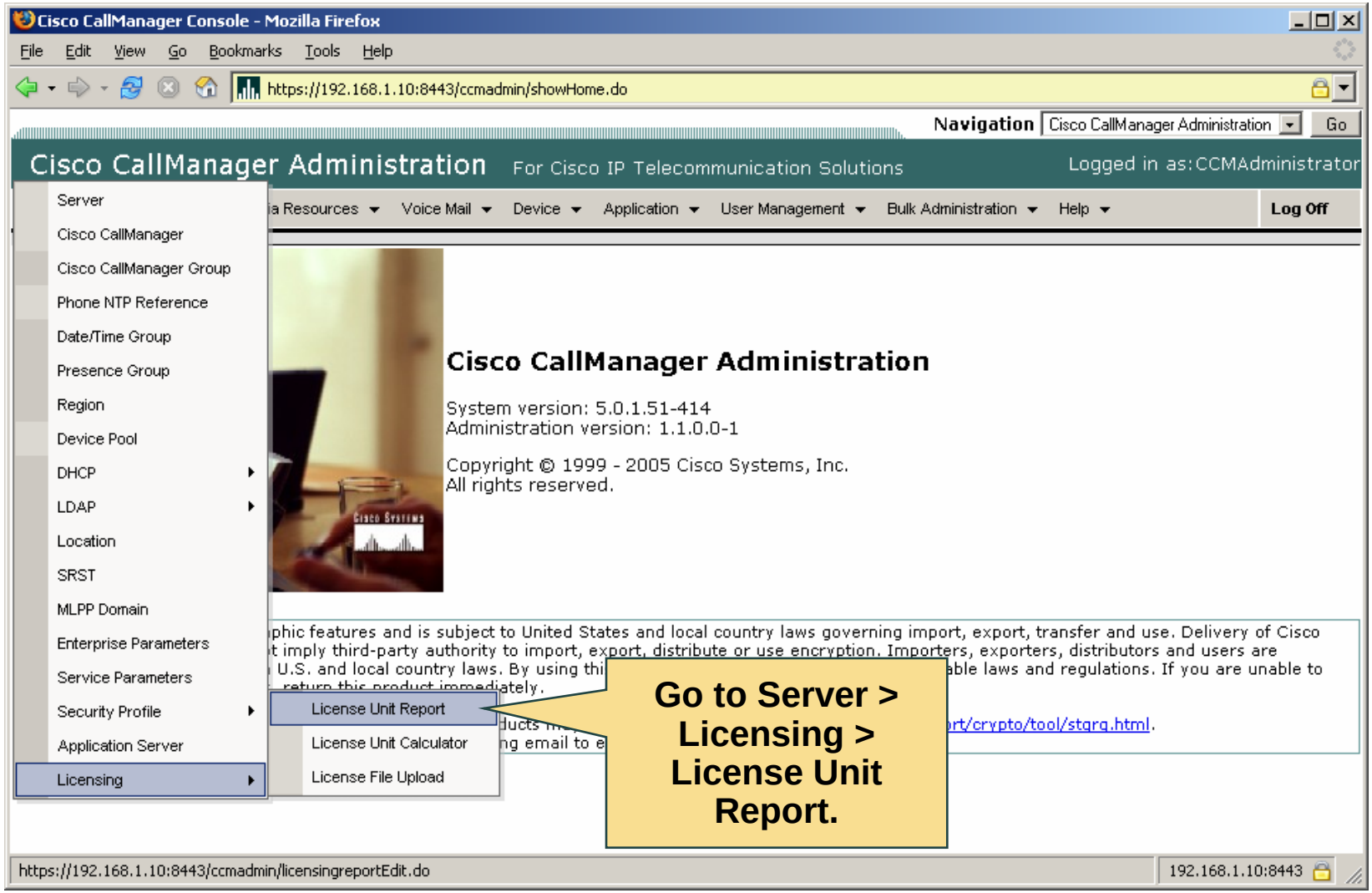
Install License File: Verify License Status



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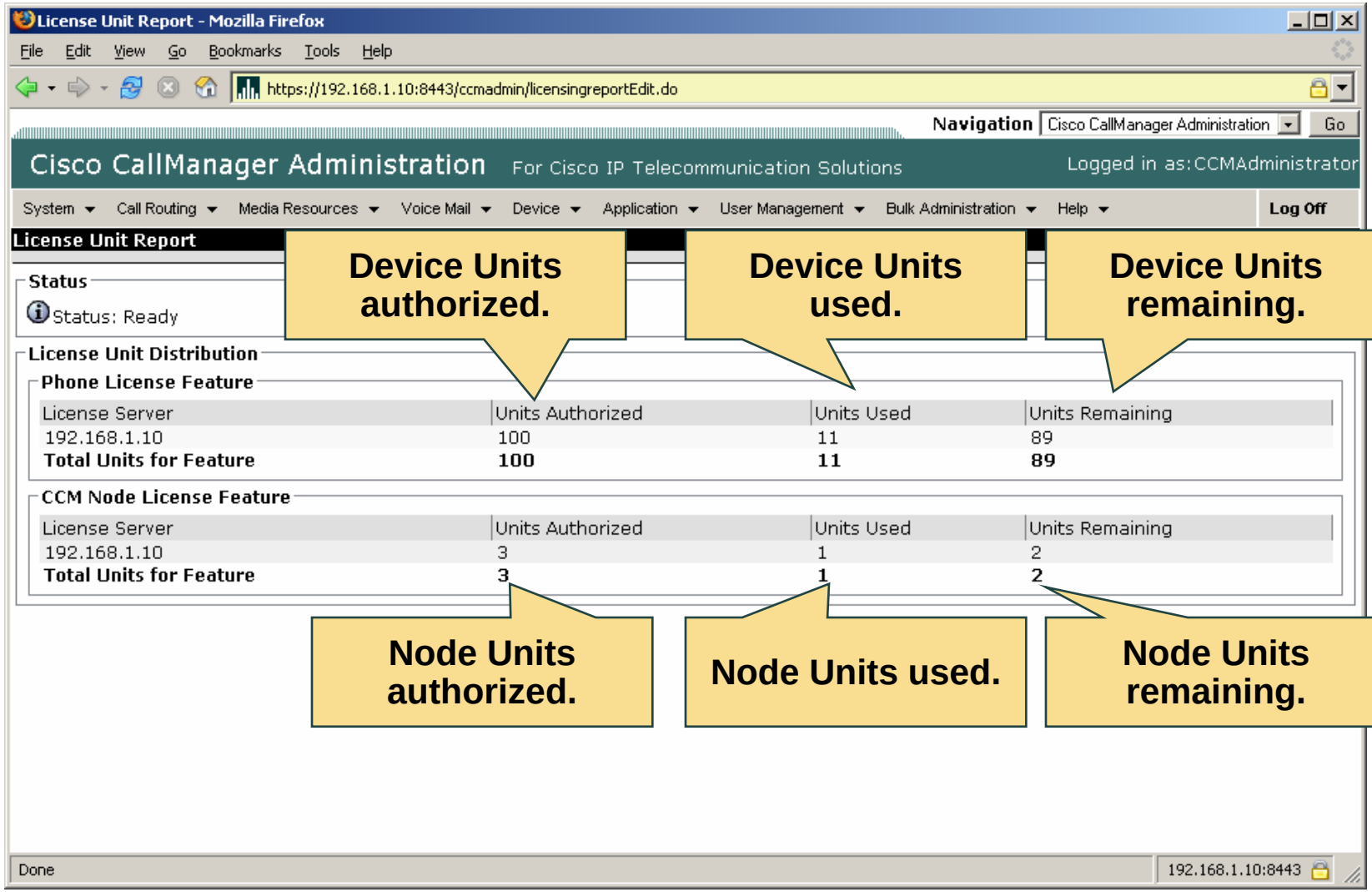
Install License File: Verify License Status



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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.

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Deploying a Cisco Unified CallManager 5.0 Cluster

Upgrading to Cisco Unified CallManager 5.0

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Objectives

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

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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**

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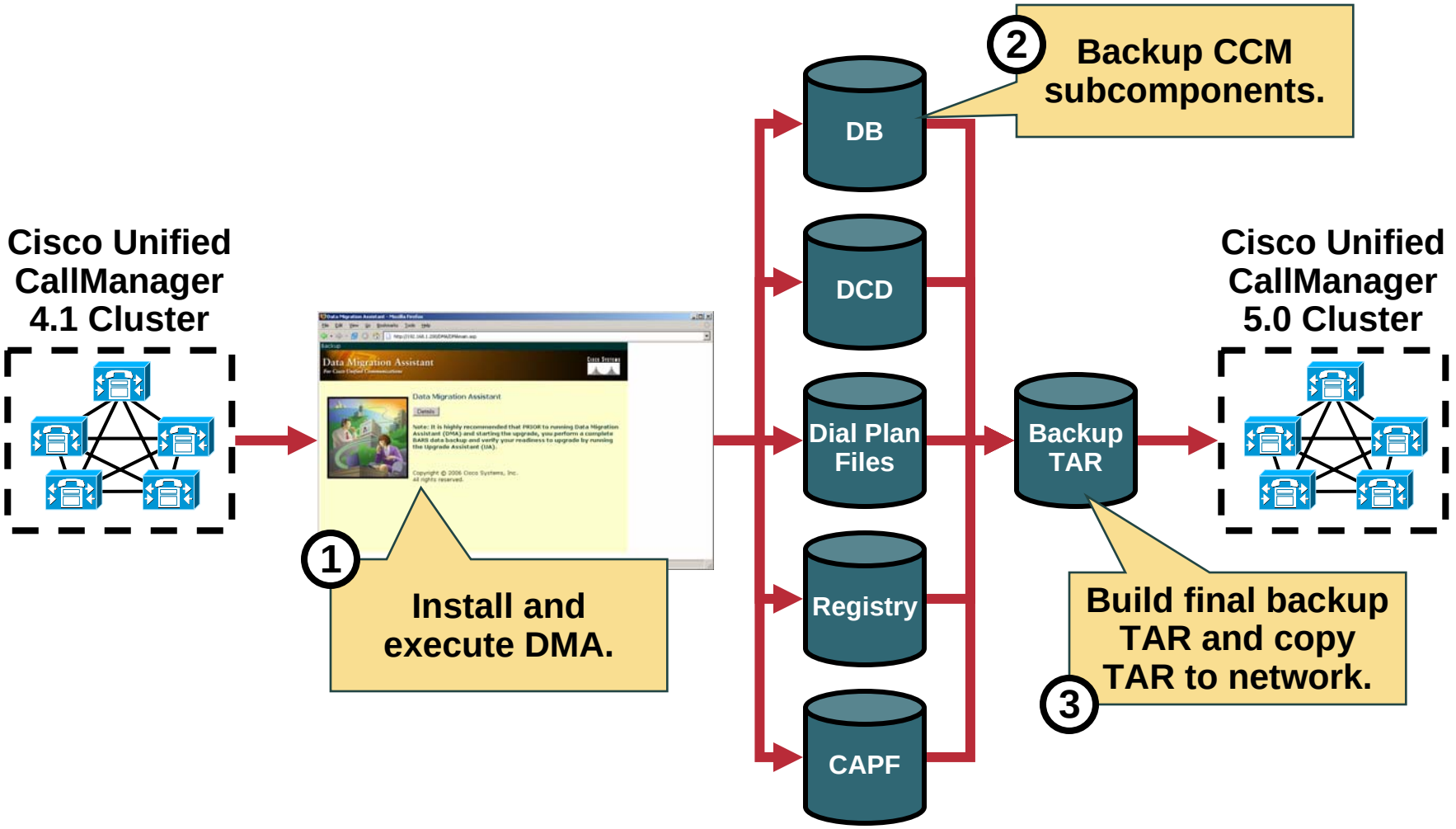
Data Migration Assistant Overview



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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

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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.**

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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).

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DMA Back-Up Procedure: Access DMA

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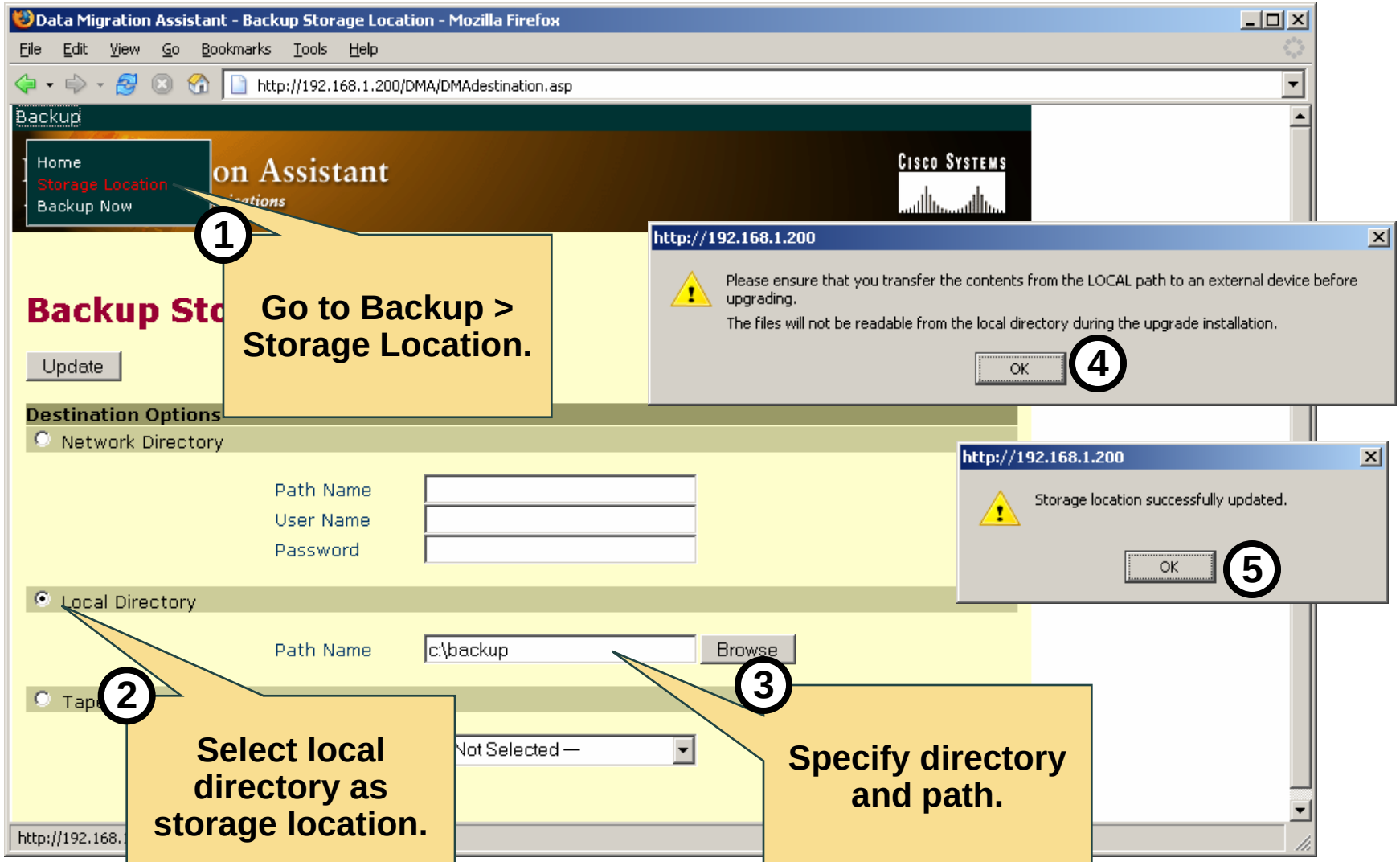
DMA Back-Up Procedure: Specify Storage Location



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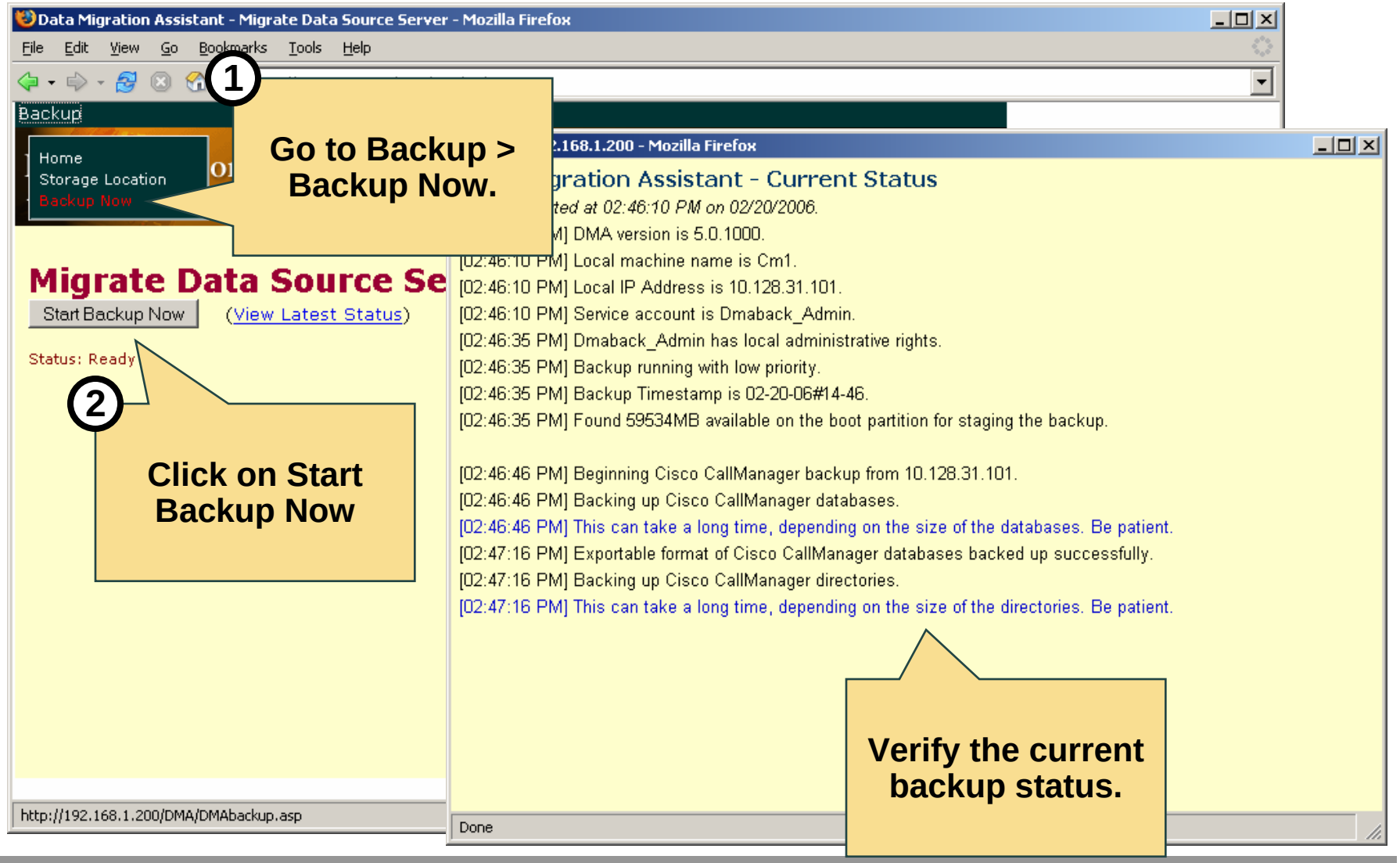
DMA Back-Up Procedure: Perform Backup Procedure



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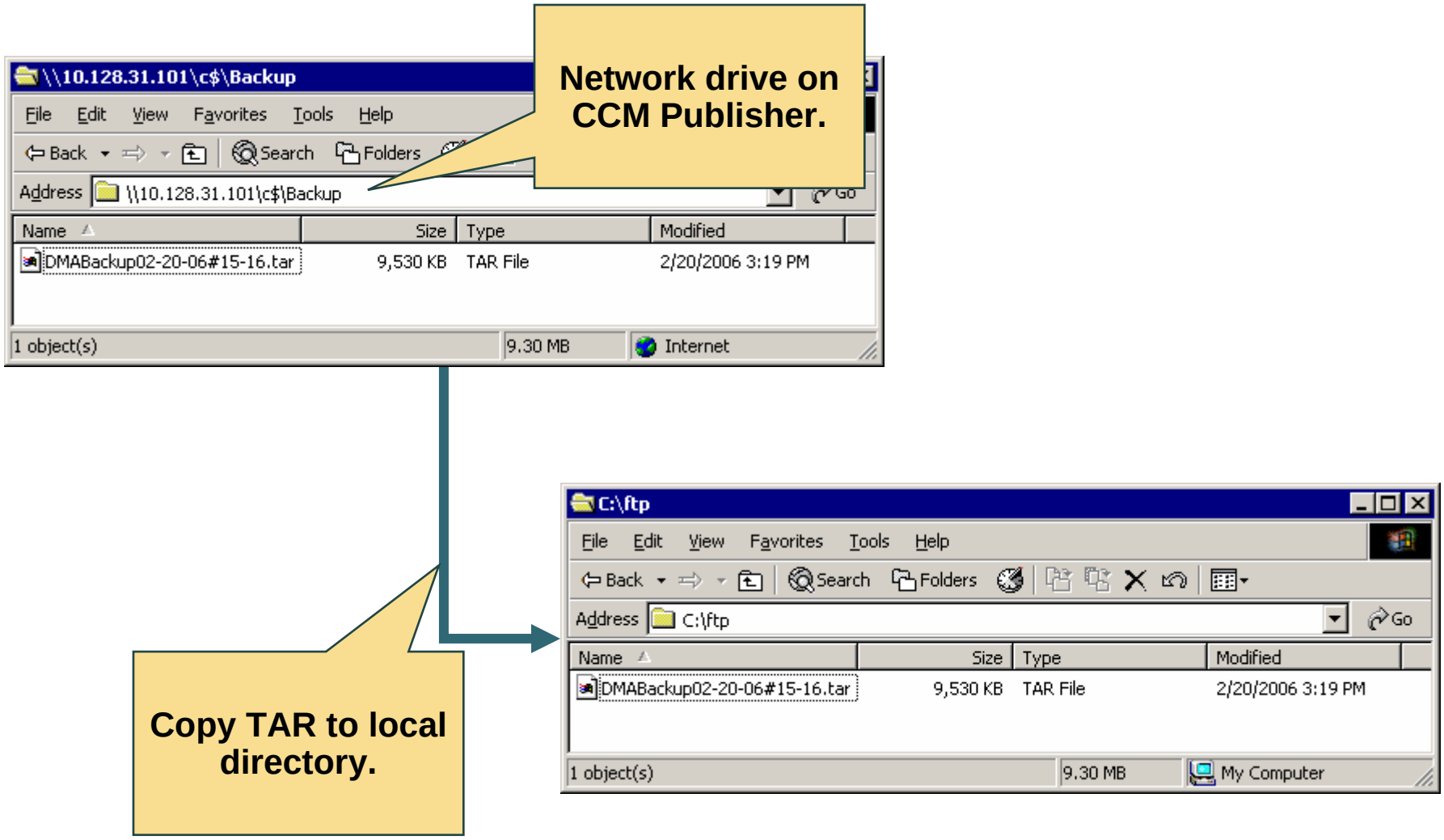
DMA Back-Up Procedure: Copy TAR



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Upgrade Procedure

Perform installation procedure using the upgrade option:

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

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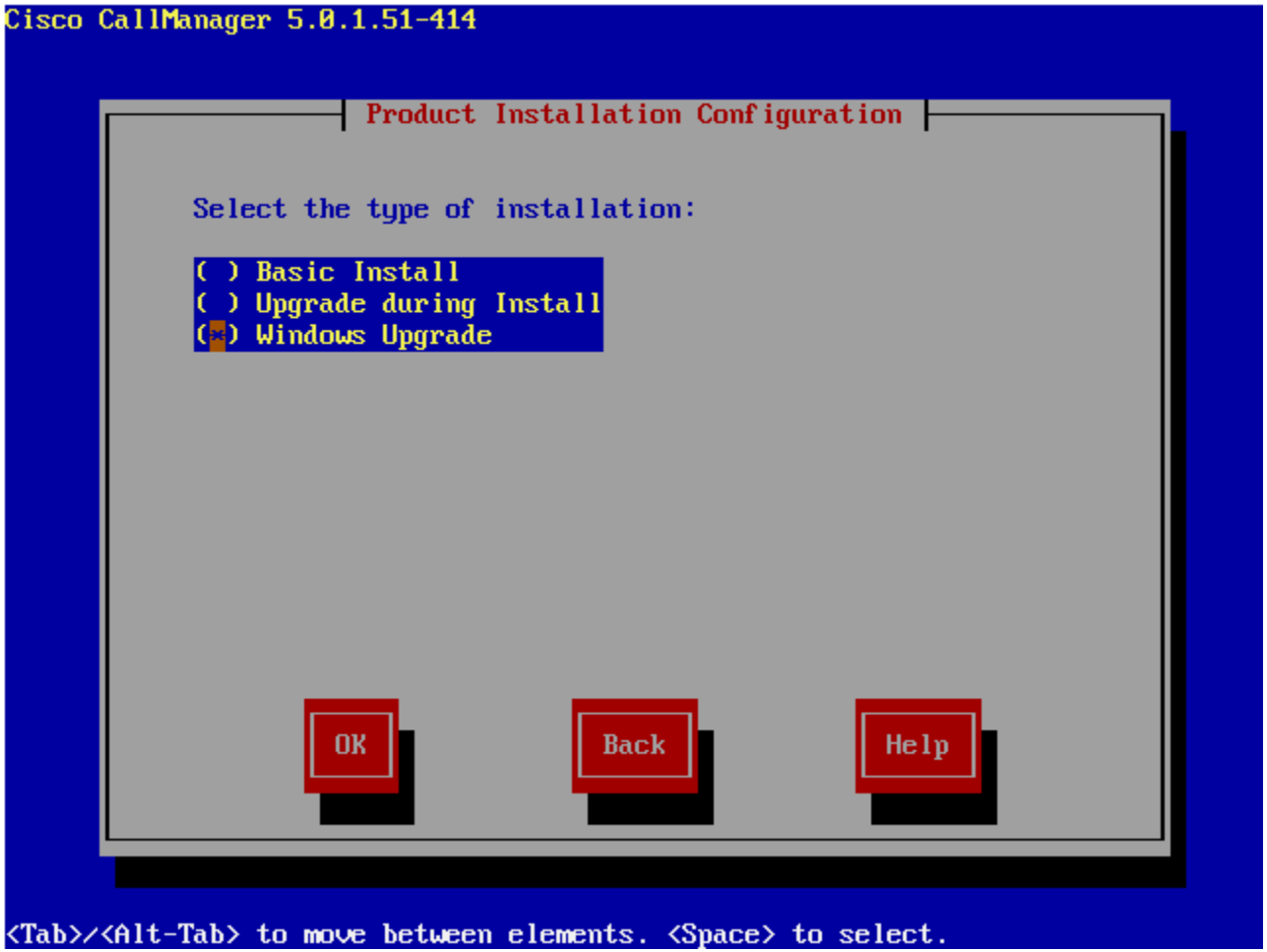
Upgrade Procedure



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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.

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Administering Cisco Unified CallManager 5.0

Performing General Administration

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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

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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.

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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>**

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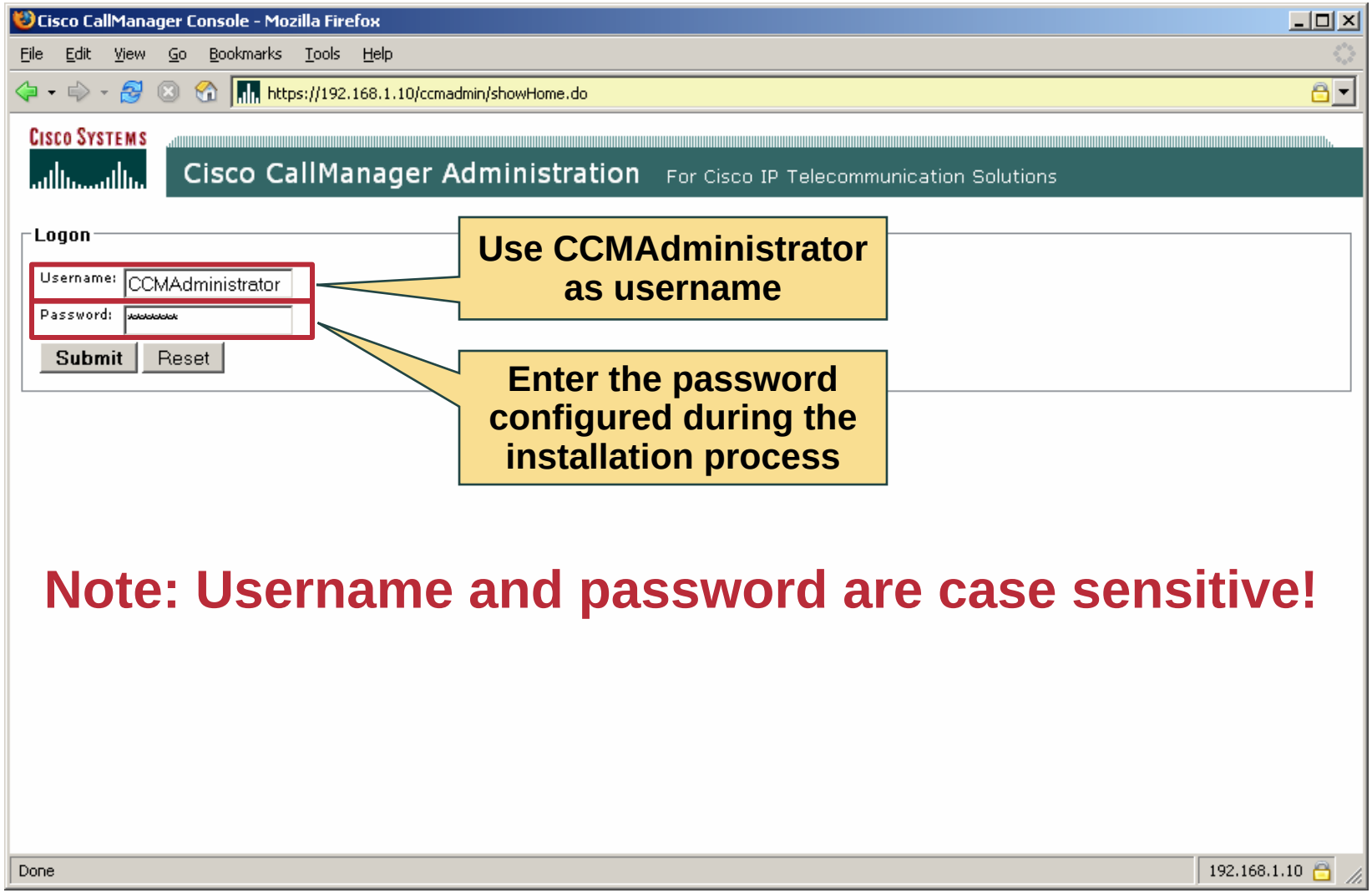
Accessing Cisco Unified CallManager 5.0



Note: Username and password are case sensitive!

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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.**

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Cisco Unified CallManager 5.0 System Menu: Server Parameters

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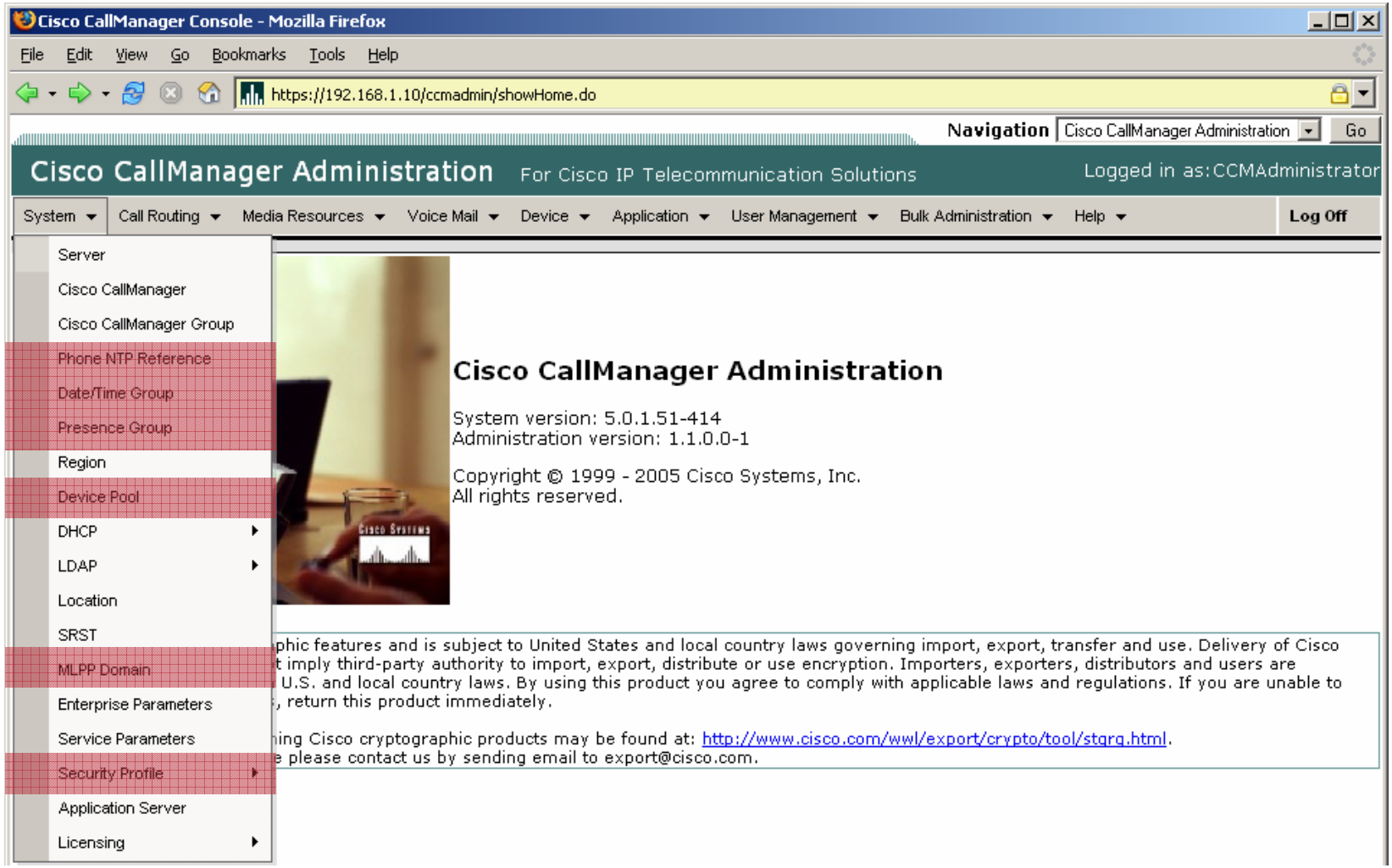
Cisco Unified CallManager 5.0 System Menu: Device Settings



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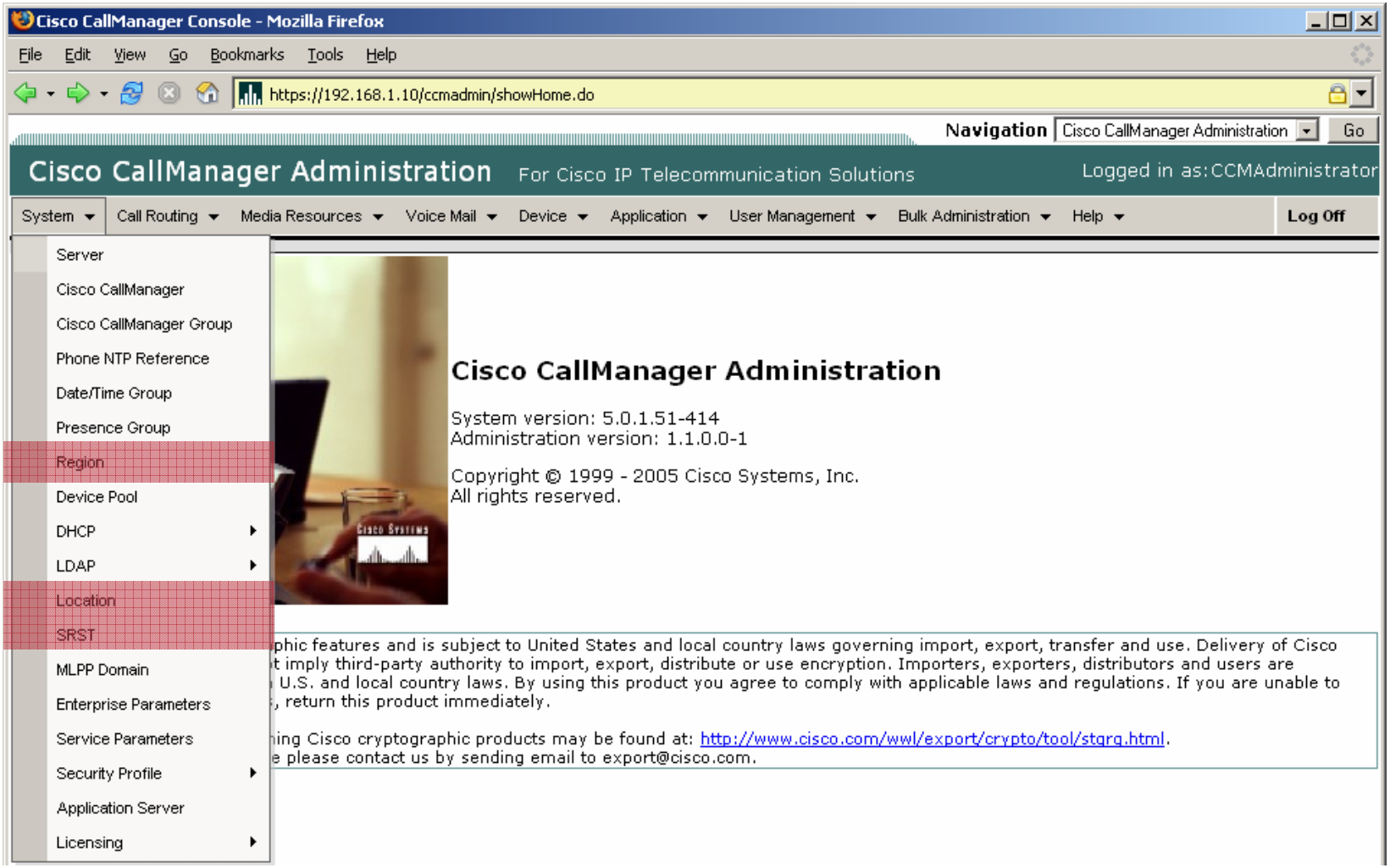
Cisco Unified CallManager 5.0
System Menu: Location Based Settings



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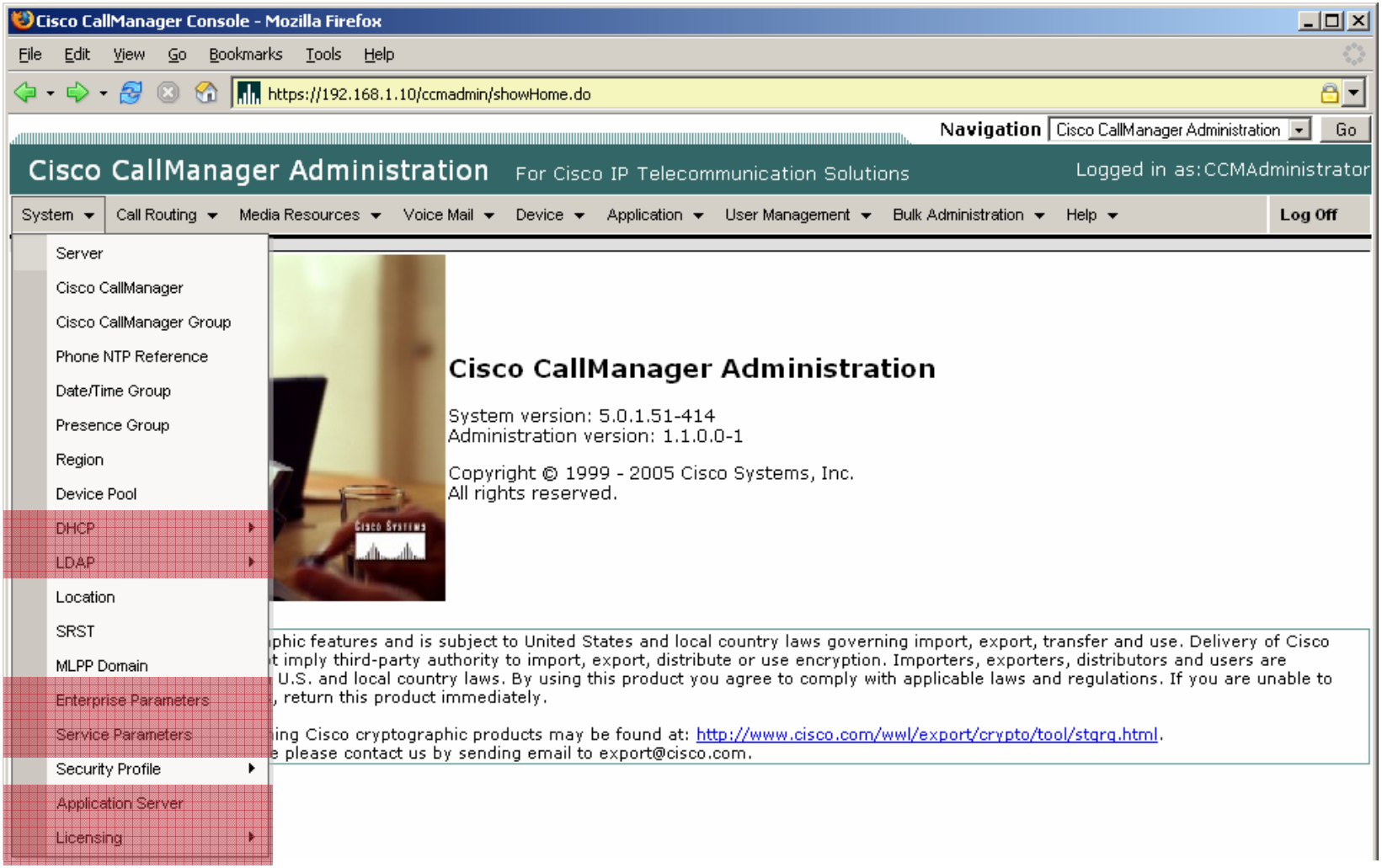
Cisco Unified CallManager 5.0 System Menu: Cluster Settings



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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.**

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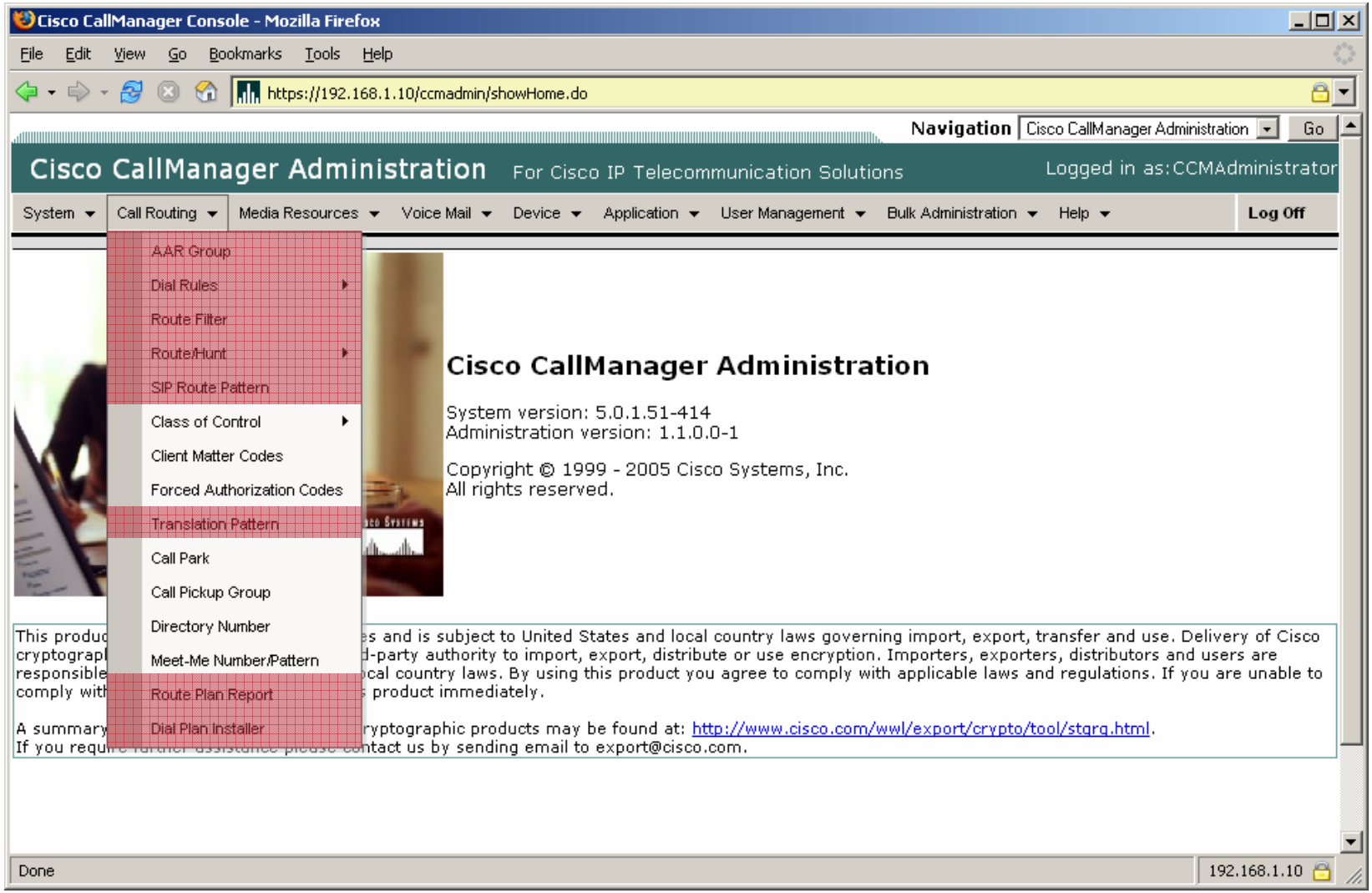
Cisco Unified CallManager 5.0
Call Routing Menu: Dial Plan



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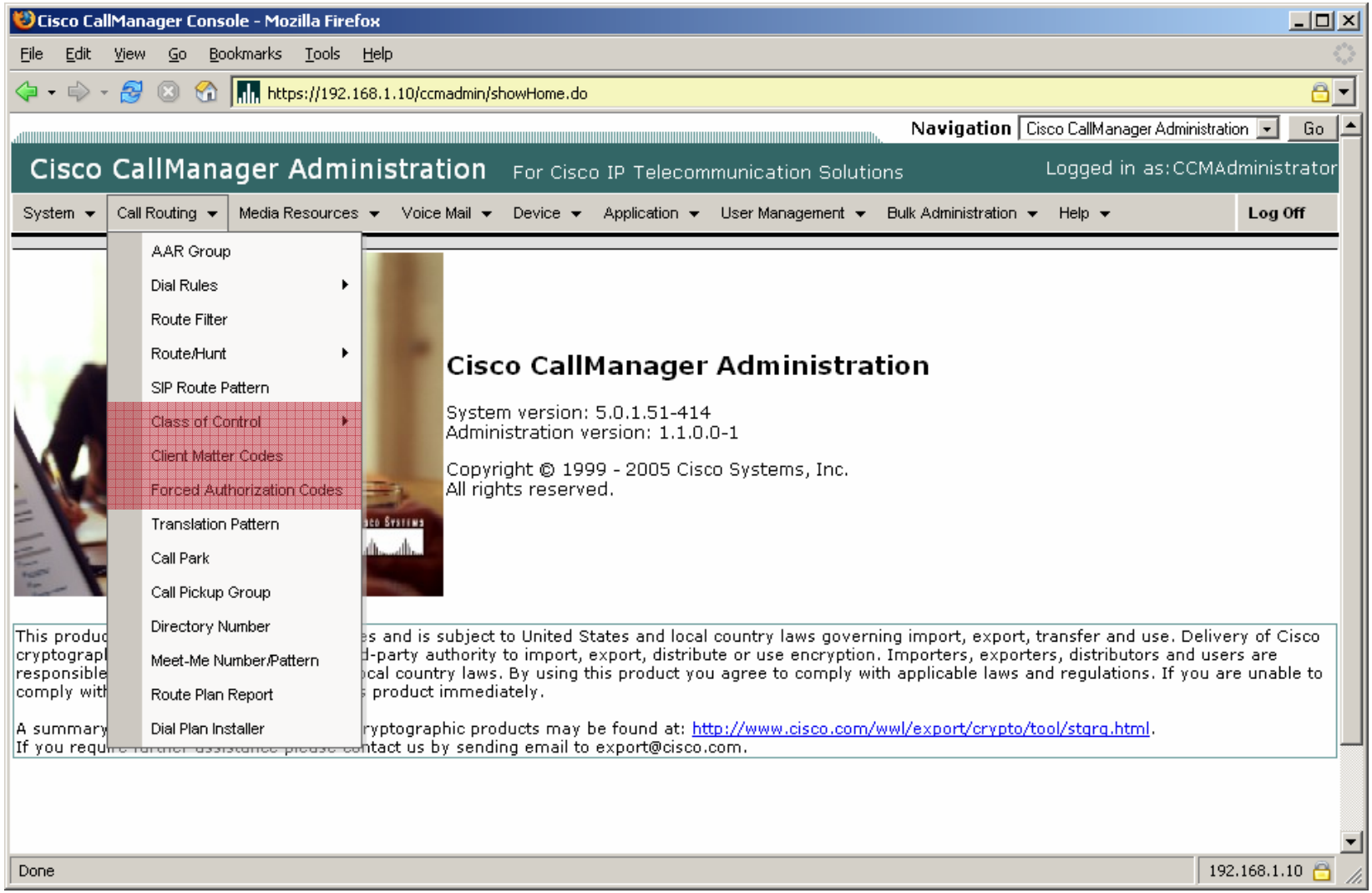
Cisco Unified CallManager 5.0
Call Routing Menu: Permissions



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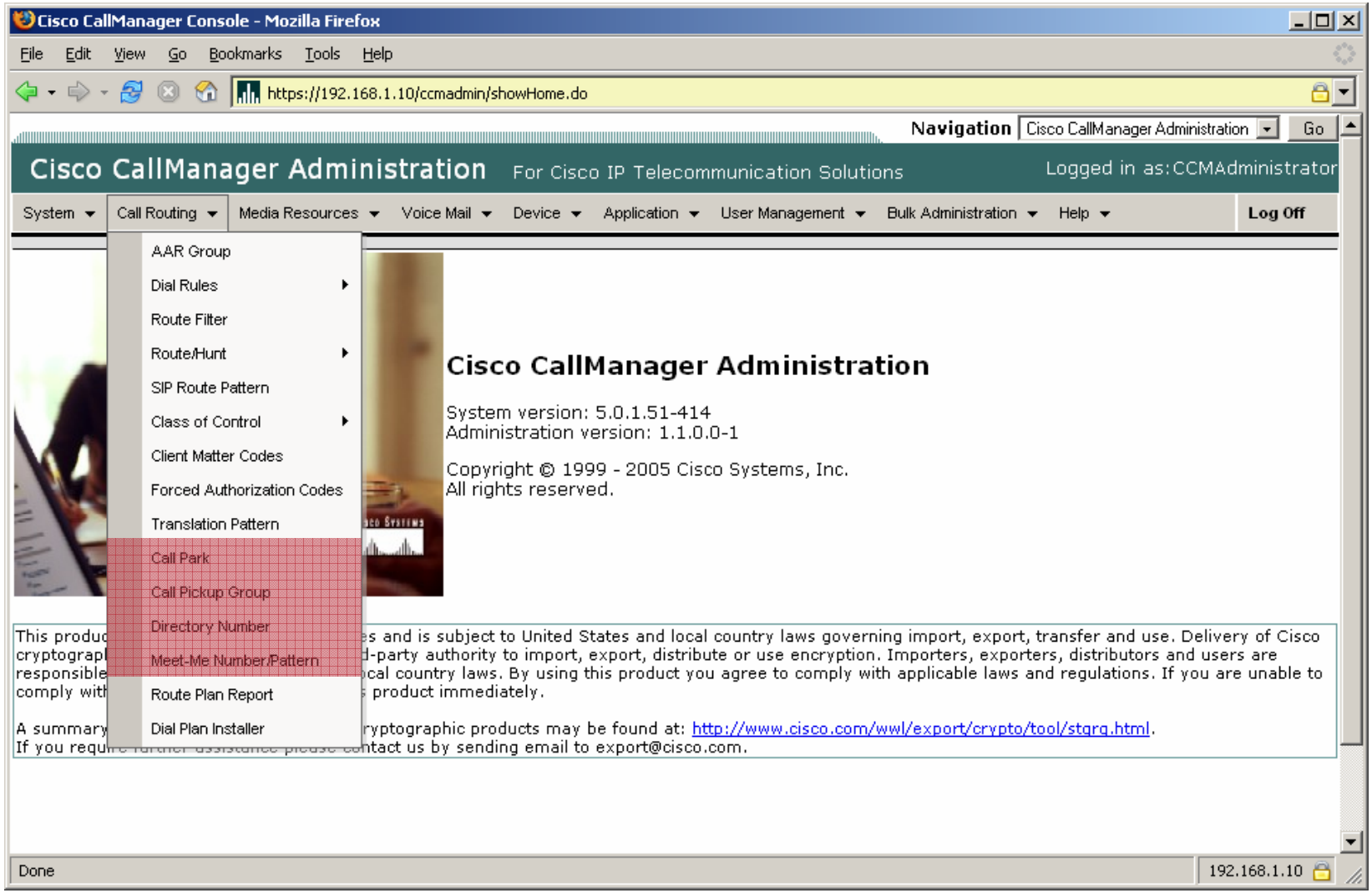
Cisco Unified CallManager 5.0
Call Routing Menu: Extension Handling



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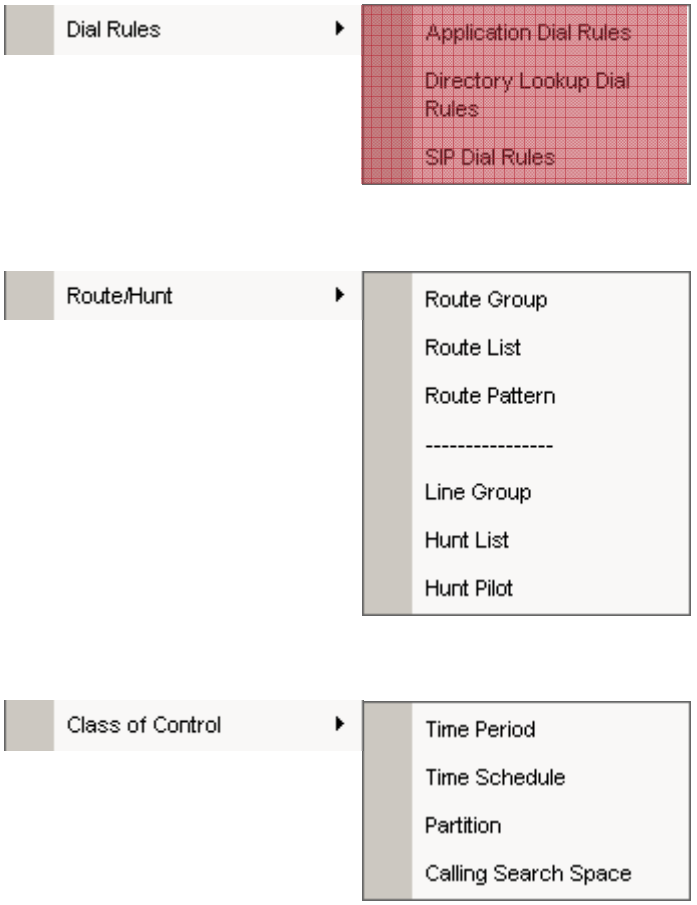
Cisco Unified CallManager 5.0 Call Routing Menu

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

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



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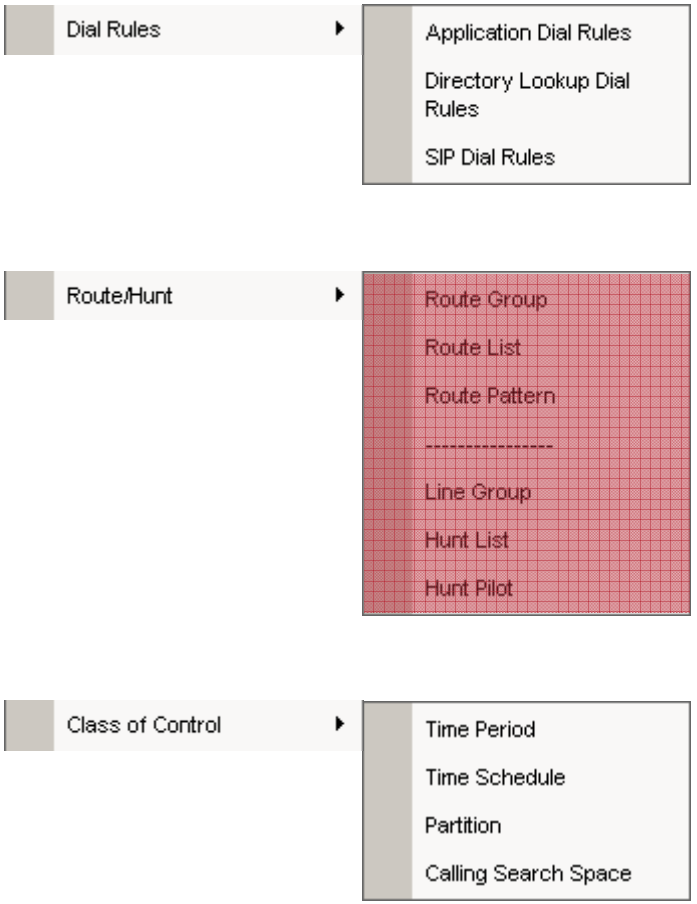
Cisco Unified CallManager 5.0 Call Routing Menu

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.**



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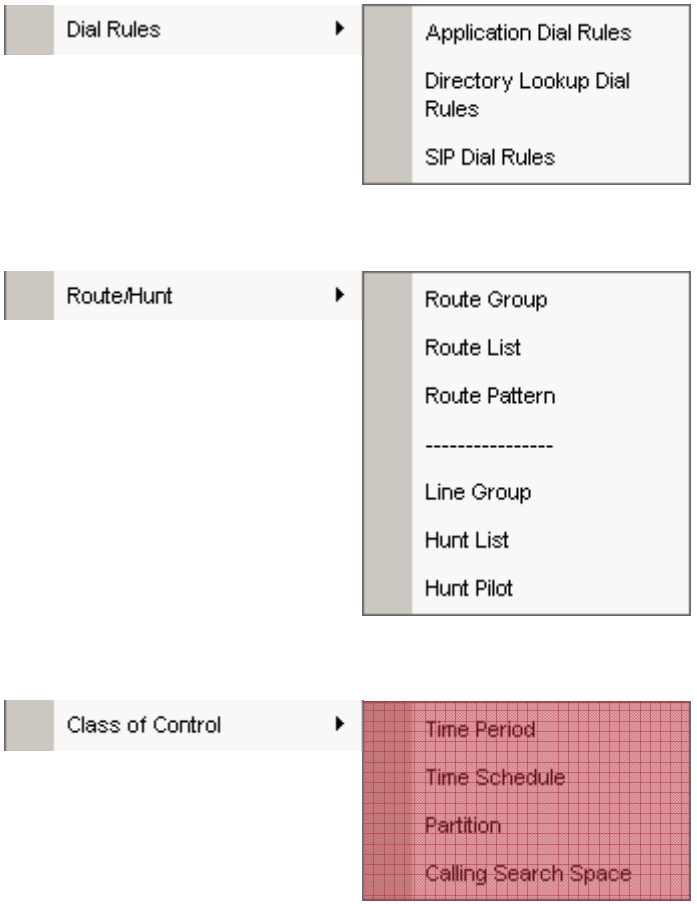
Cisco Unified CallManager 5.0 Call Routing Menu

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.**



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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.**

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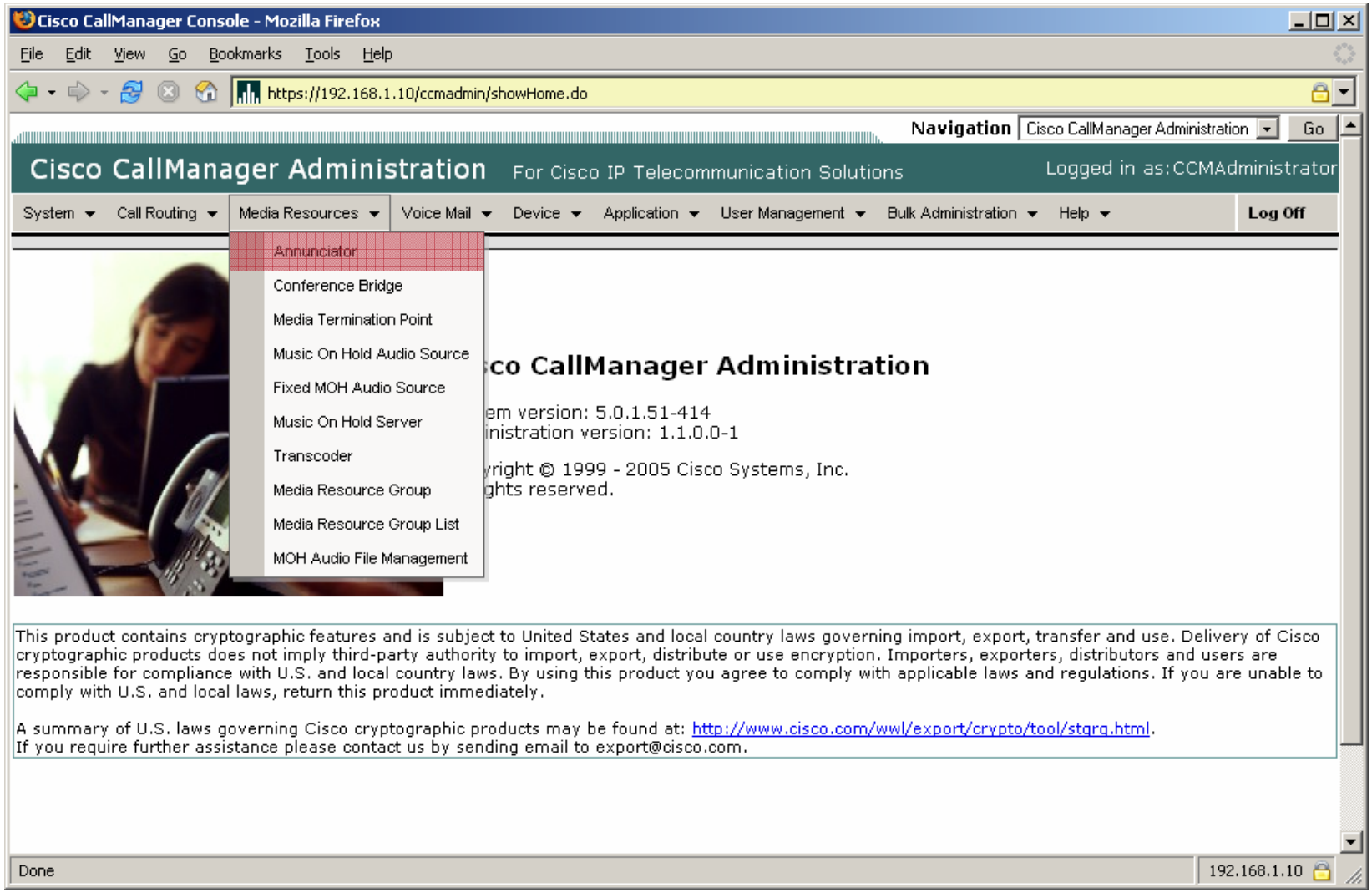
Cisco Unified CallManager 5.0 Media Resources Menu: Annunciator



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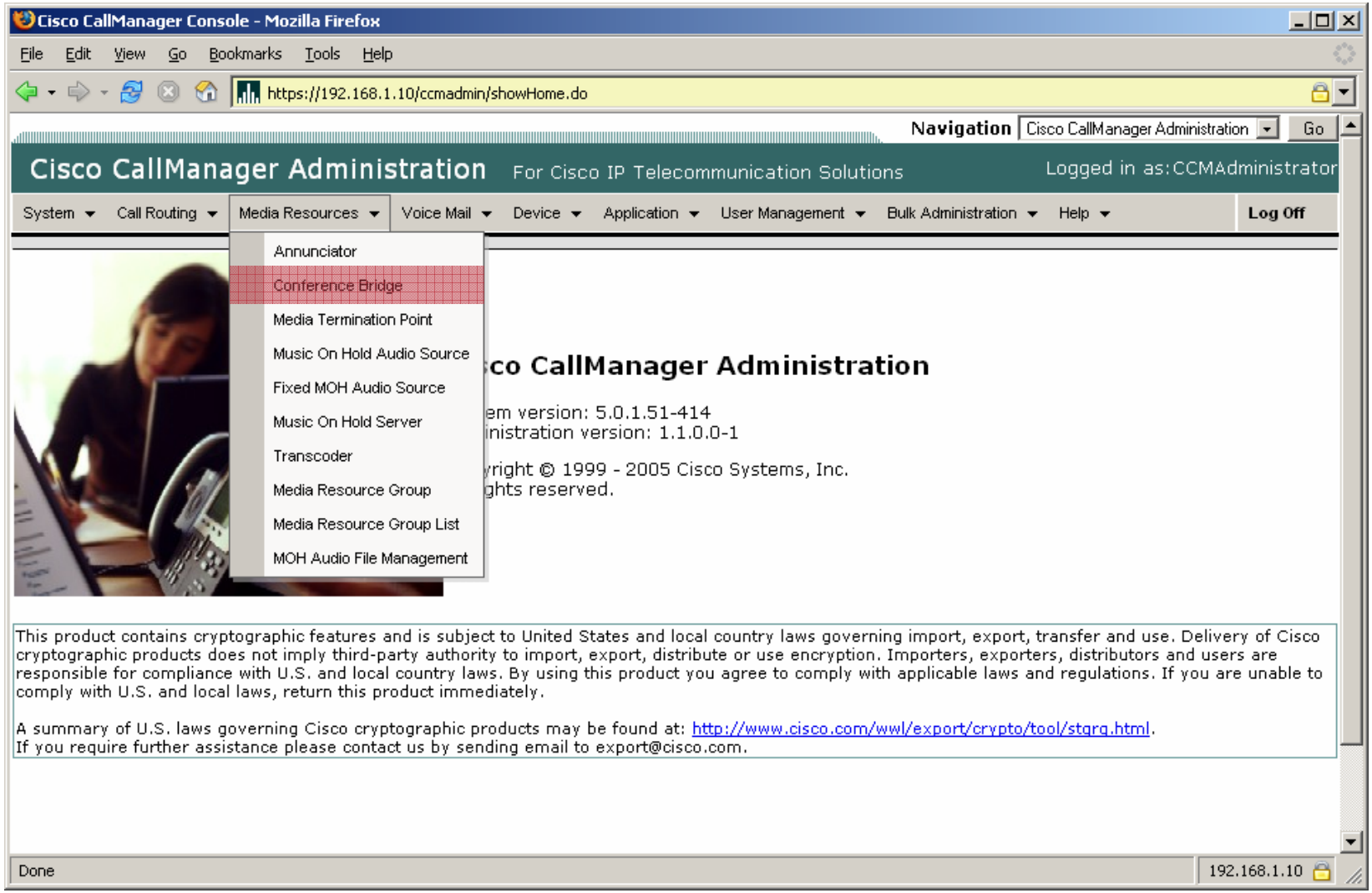
Cisco Unified CallManager 5.0 Media Resources Menu: Conference Bridge



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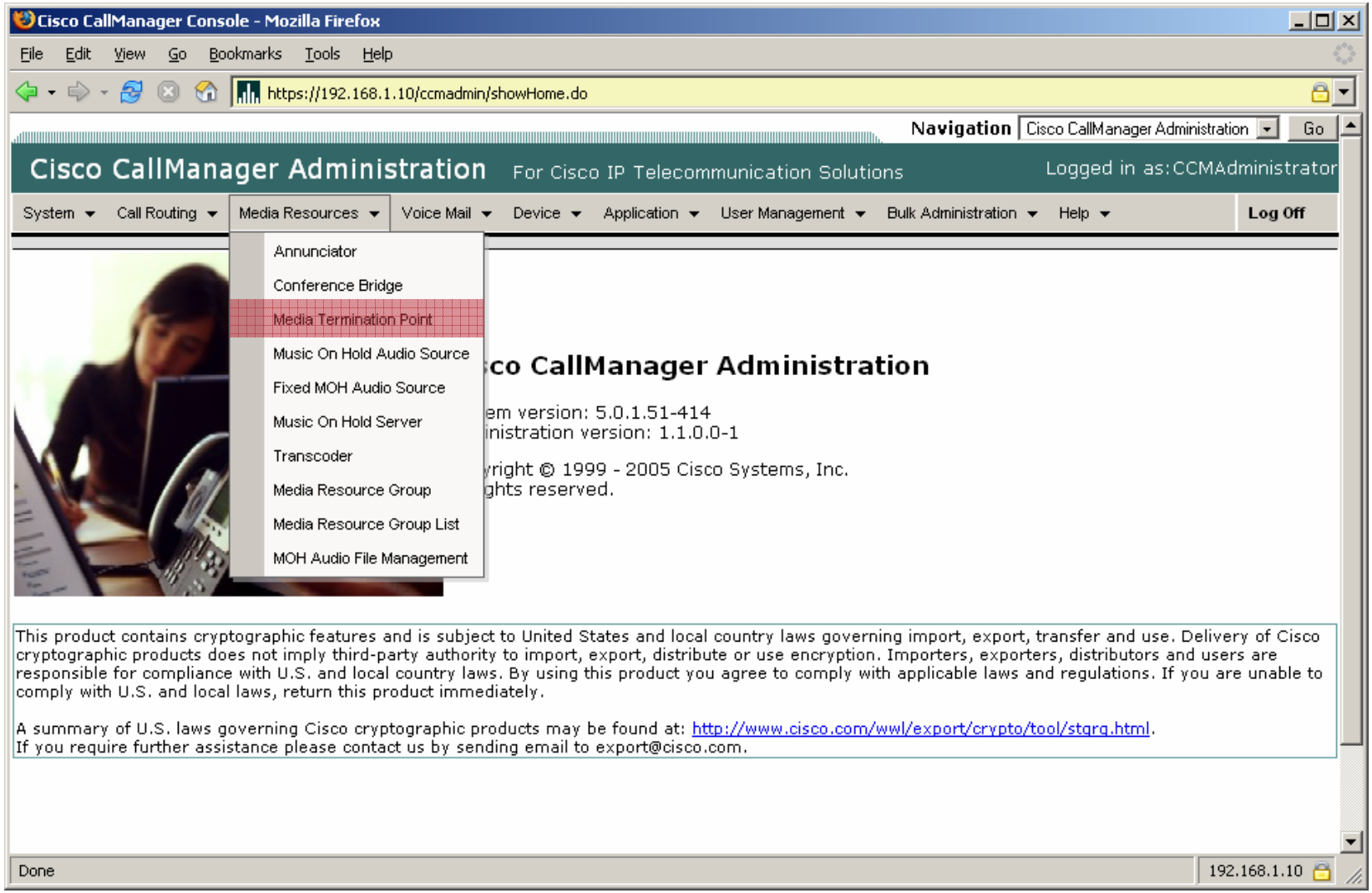
Cisco Unified CallManager 5.0 Media Resources Menu: Media Termination Point



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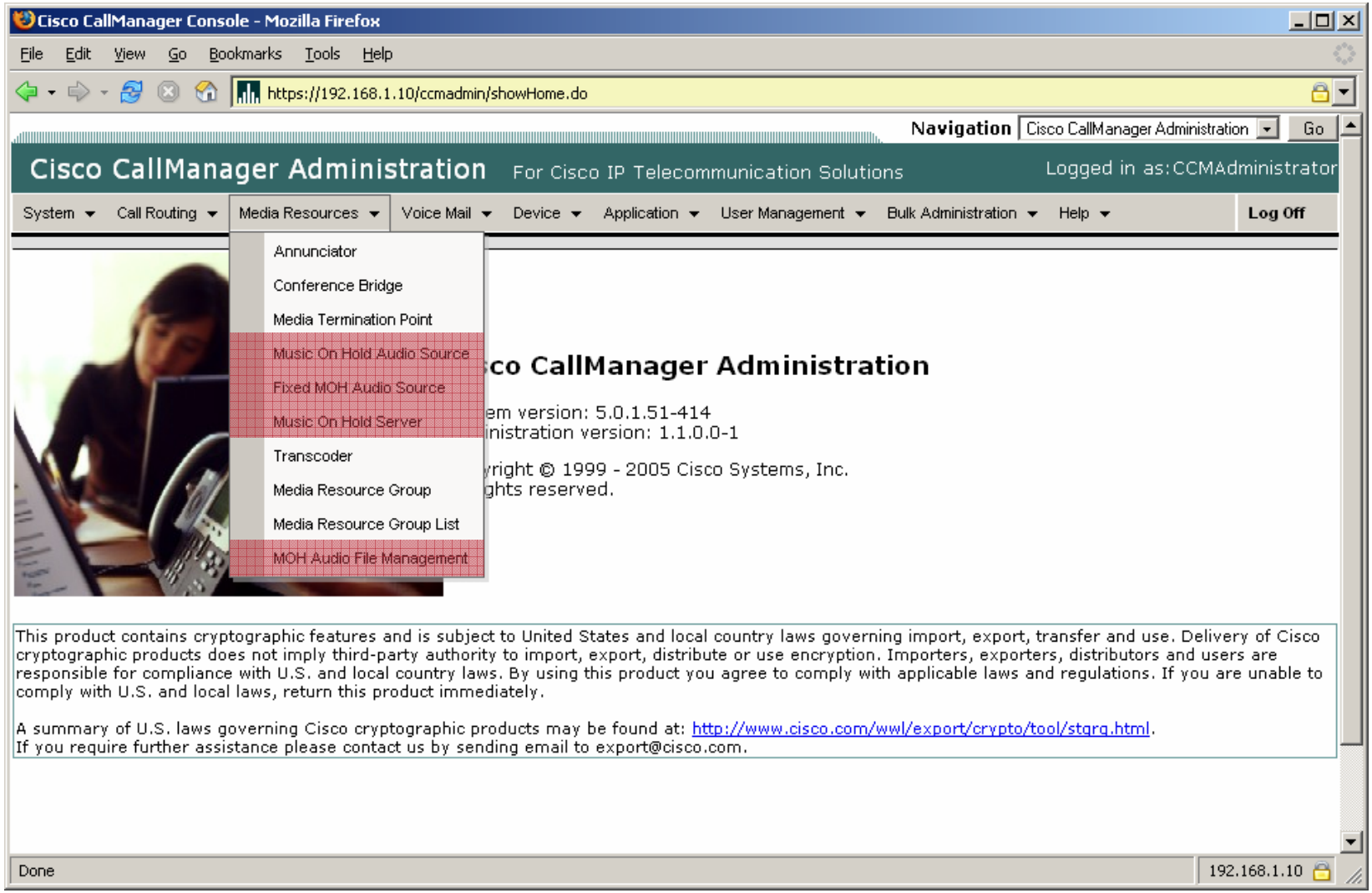
Cisco Unified CallManager 5.0 Media Resources Menu: Music On Hold



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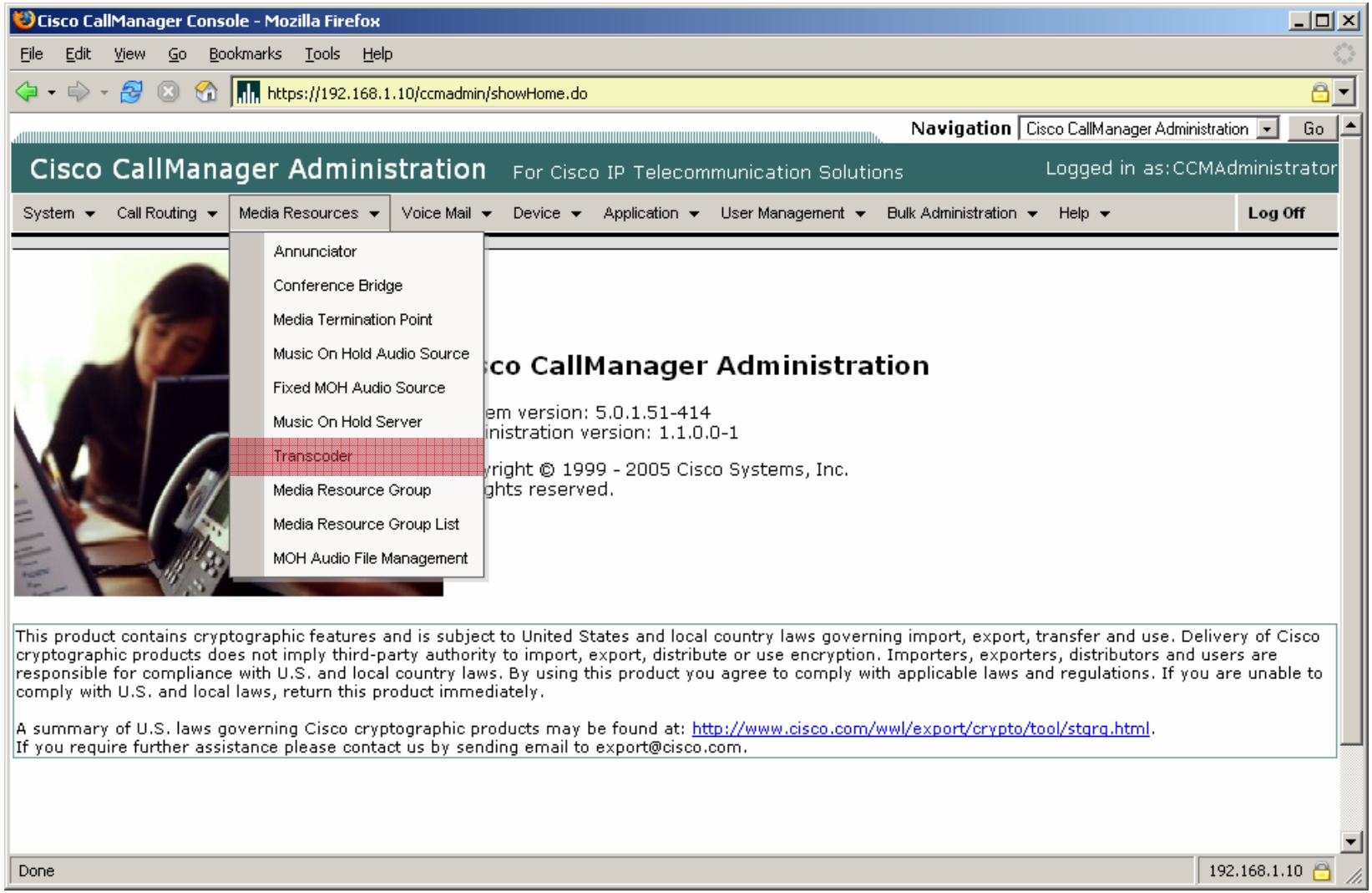
Cisco Unified CallManager 5.0 Media Resources Menu: Transcoder



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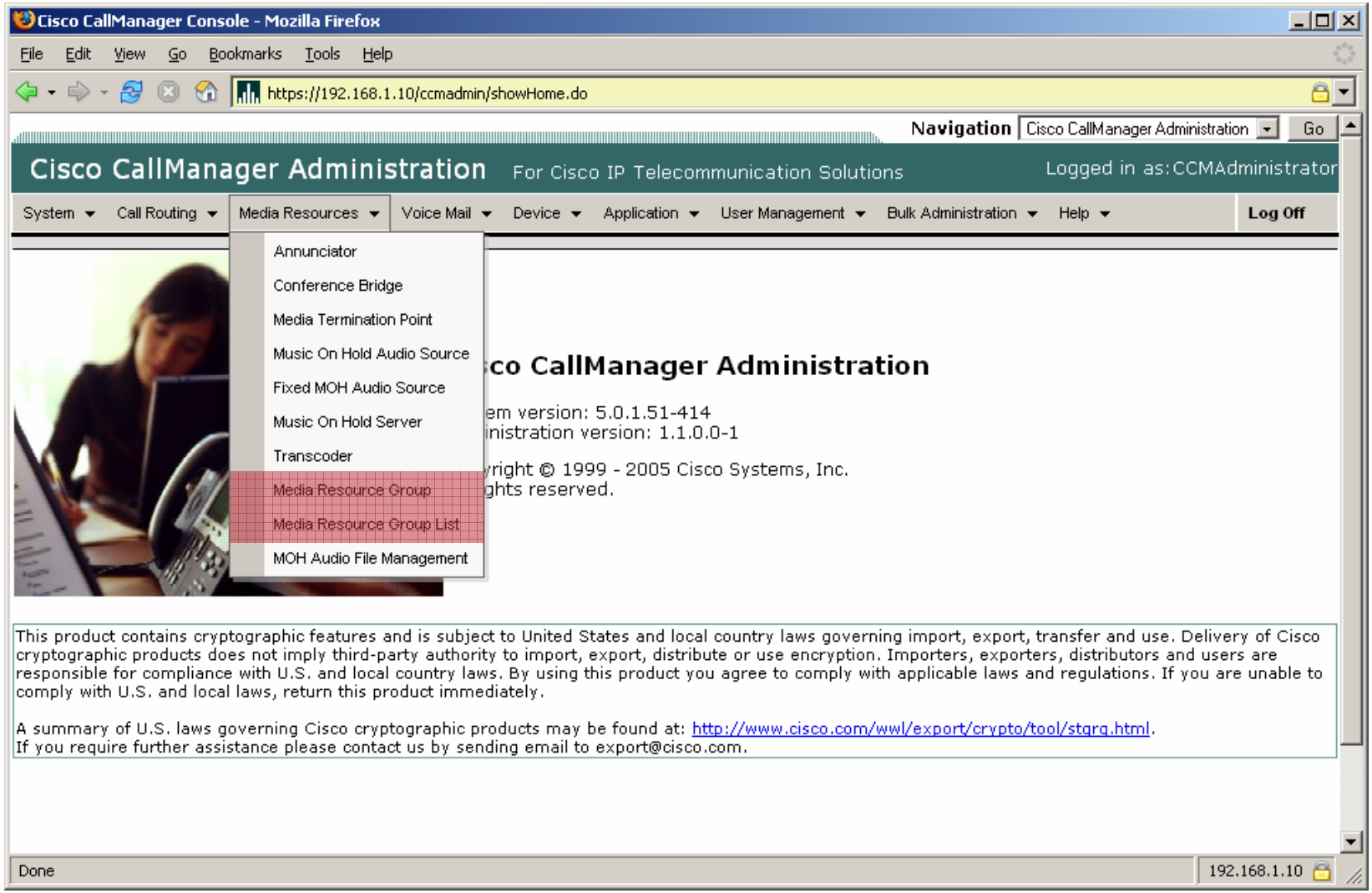
Cisco Unified CallManager 5.0 Media Resources Menu: Media Resources



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Cisco Unified CallManager 5.0
Voice Mail Menu

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

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

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Cisco Unified CallManager 5.0
Voice Mail Menu: Voice Port Configuration

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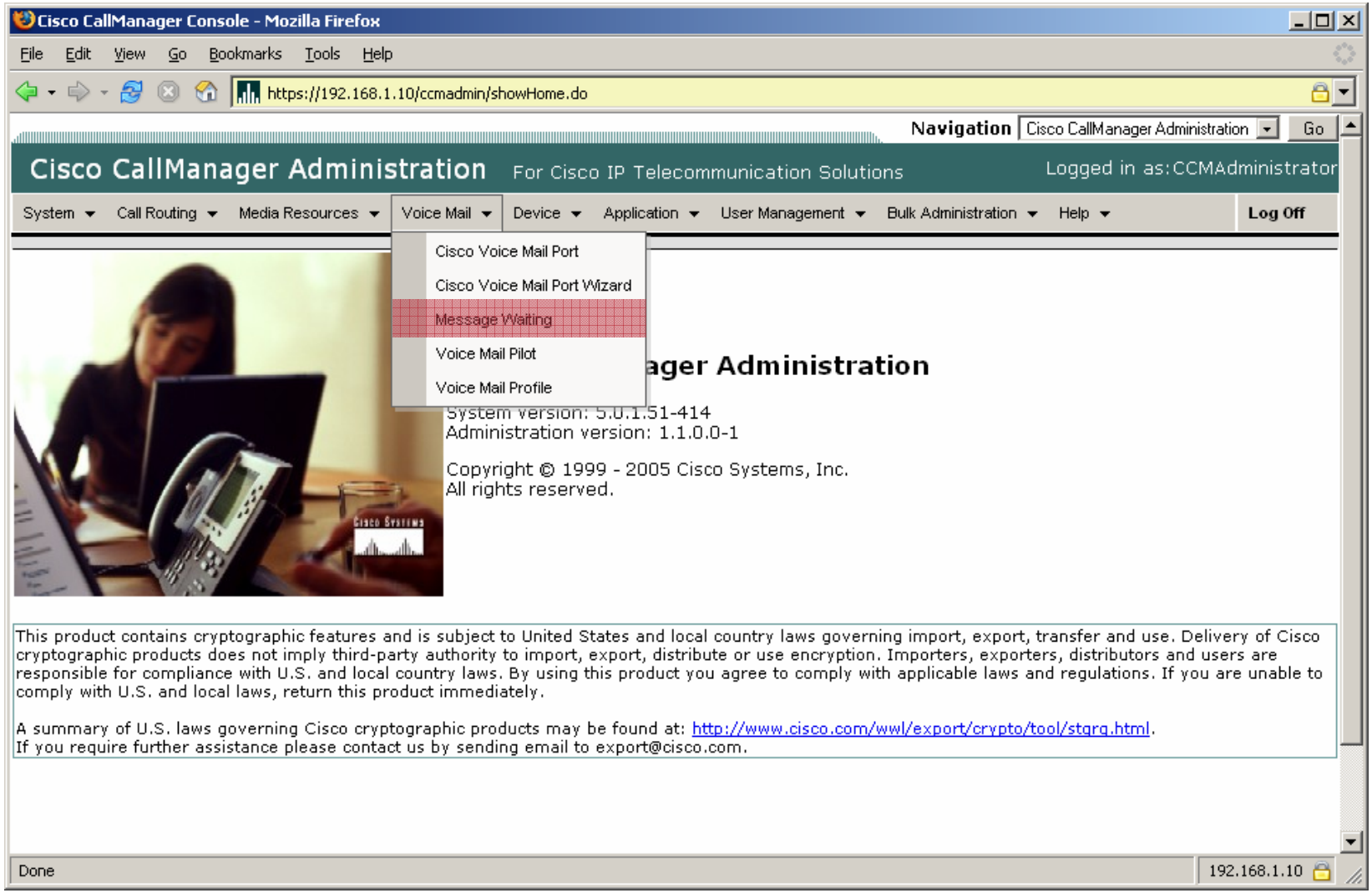
Cisco Unified CallManager 5.0 Voice Mail Menu: Message Waiting Indication



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Cisco Unified CallManager 5.0
Voice Mail Menu: Voice Mail Access

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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.**

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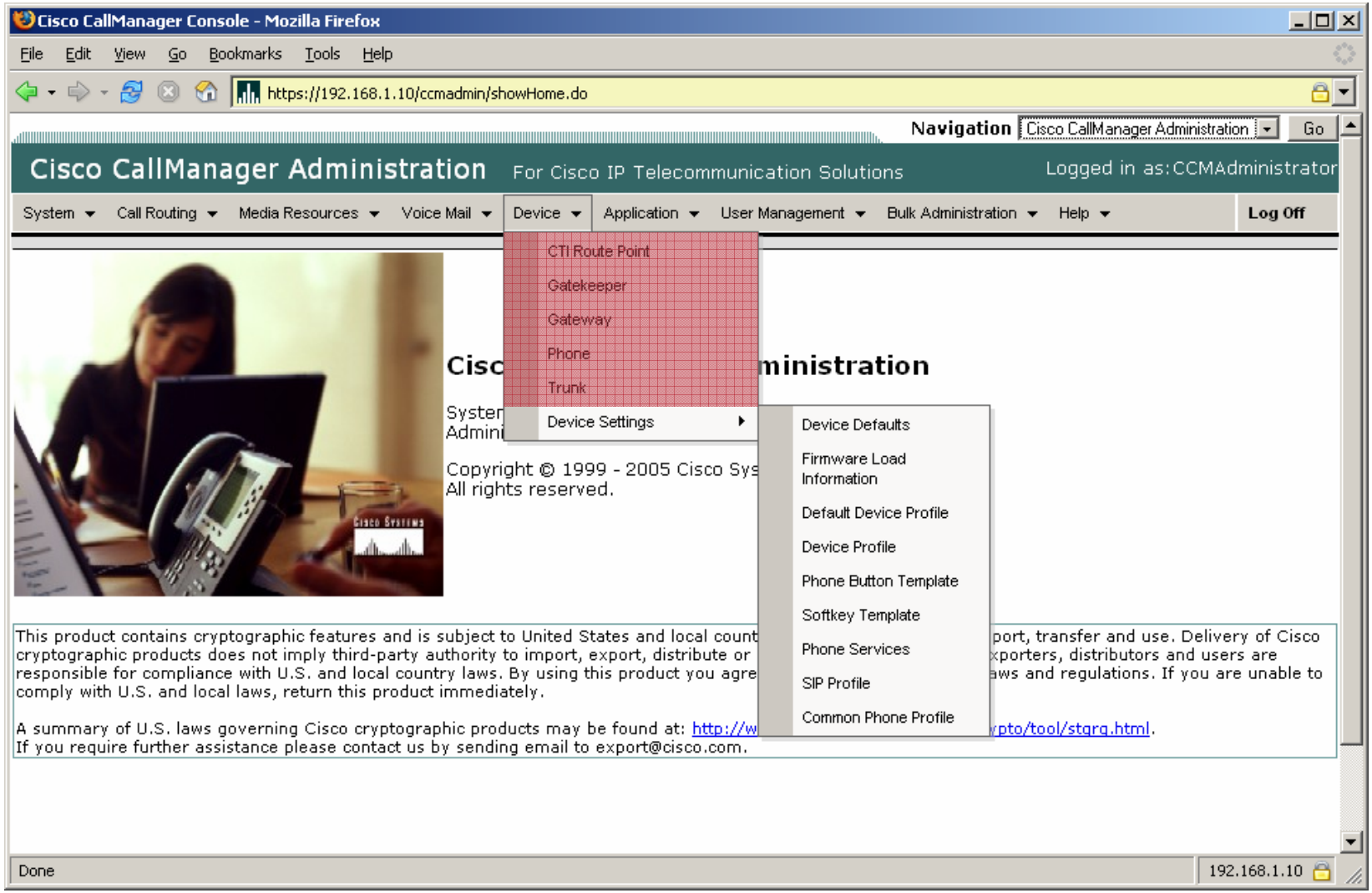
Cisco Unified CallManager 5.0
Device Menu: Physical and Logical Devices



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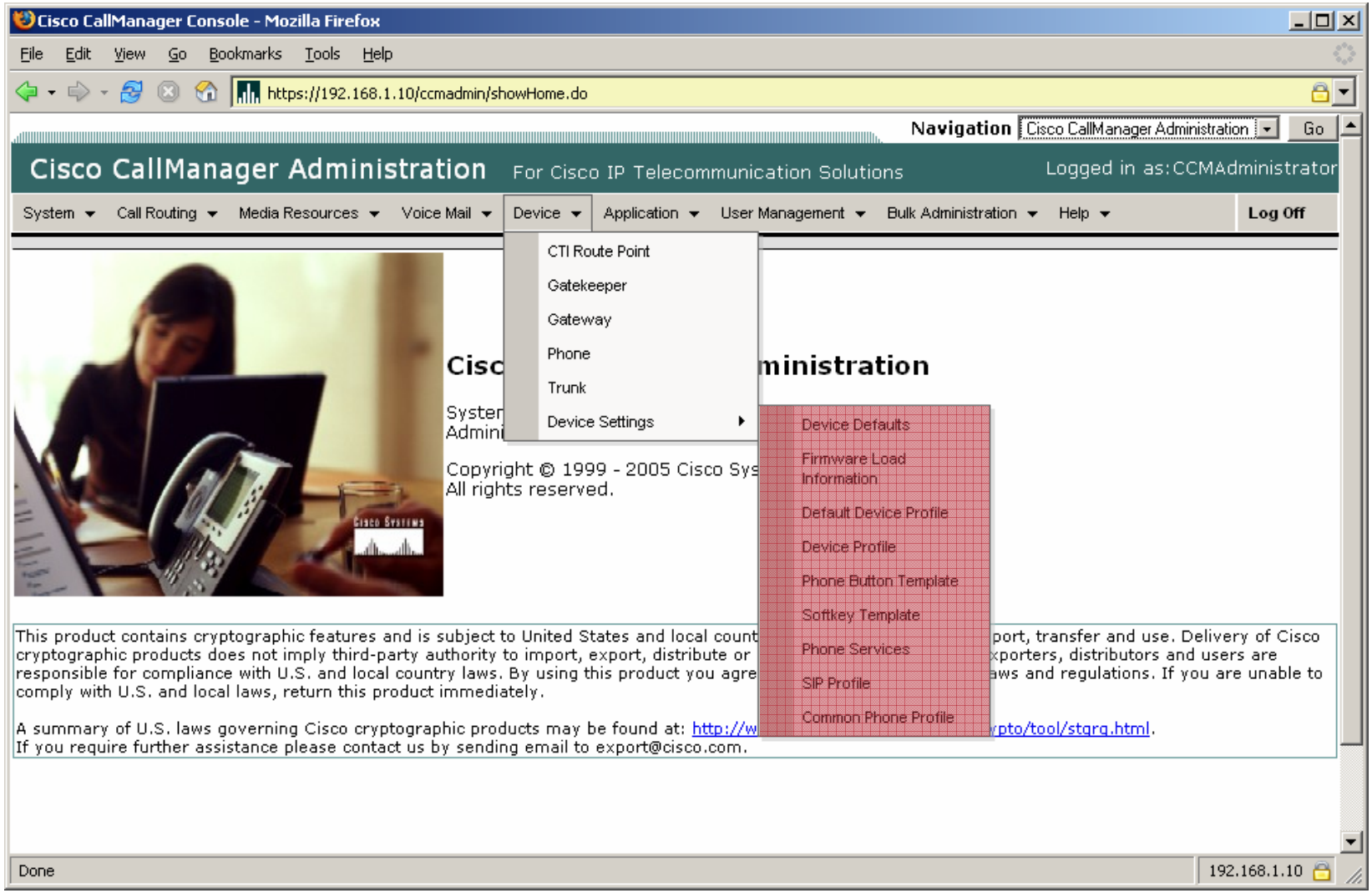
Cisco Unified CallManager 5.0 Device Menu: Profiles, Defaults, and Templates



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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.**

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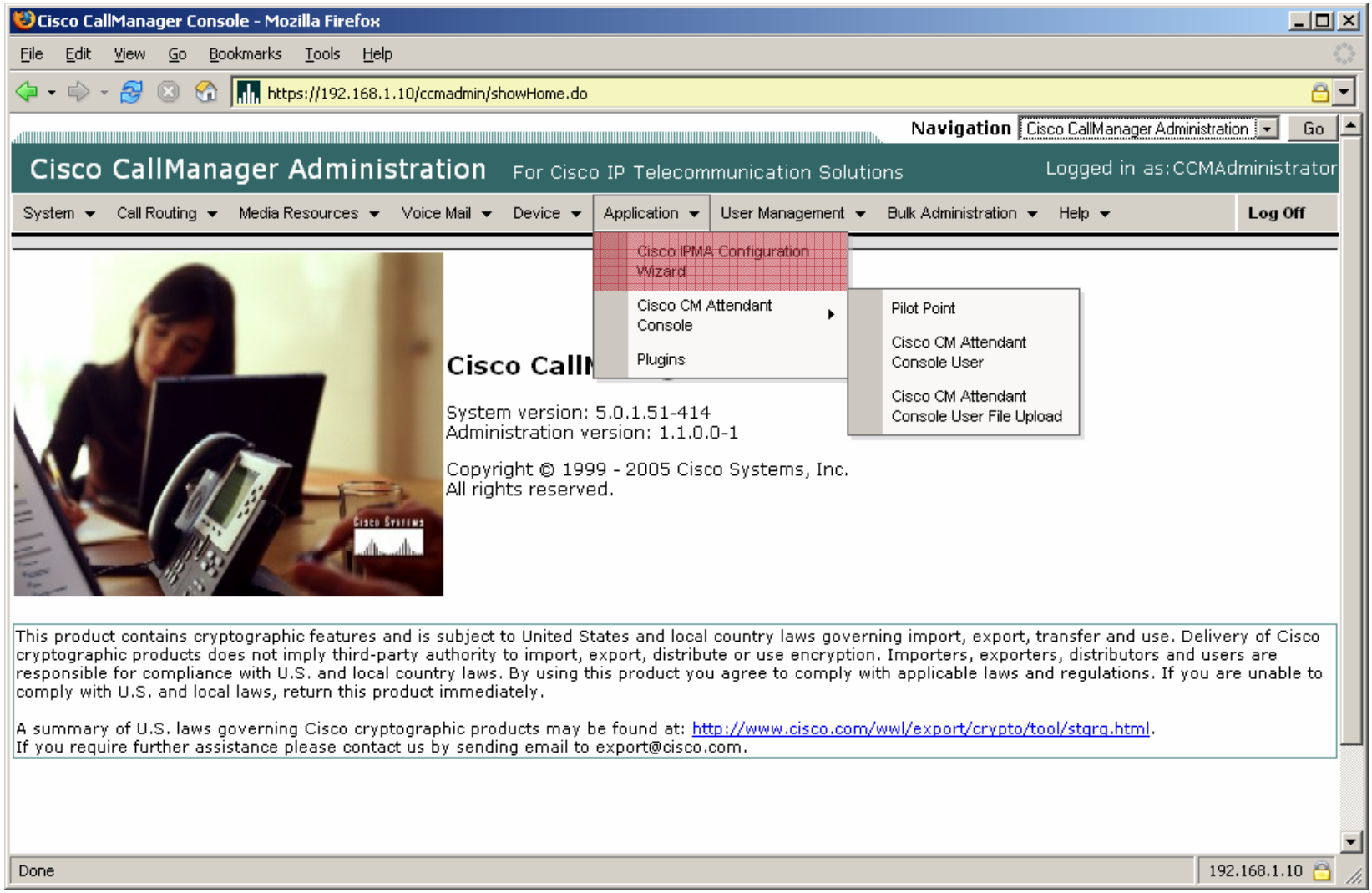
Cisco Unified CallManager 5.0 Application Menu: IPMA



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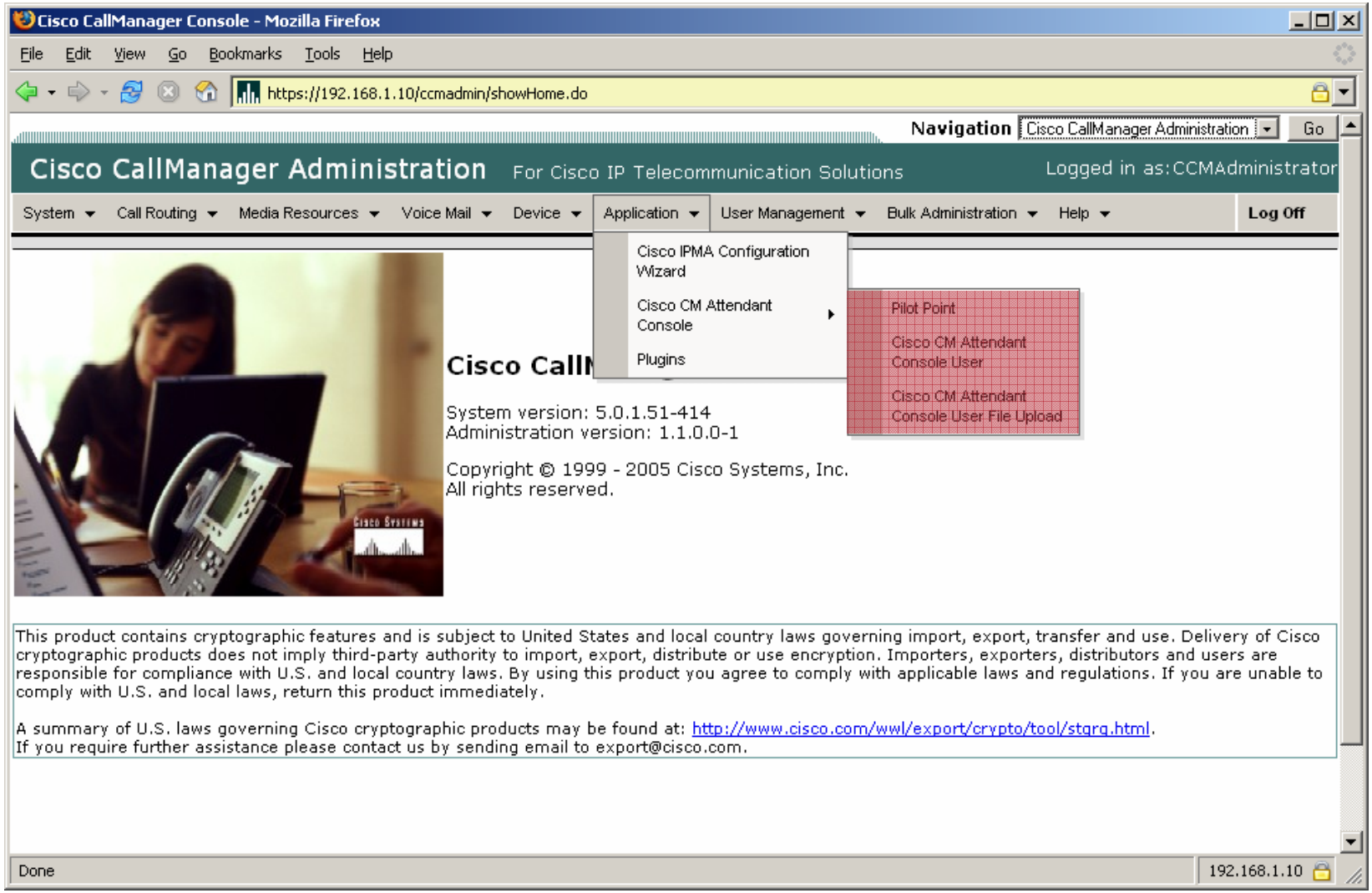
Cisco Unified CallManager 5.0 Application Menu: Attendant Console



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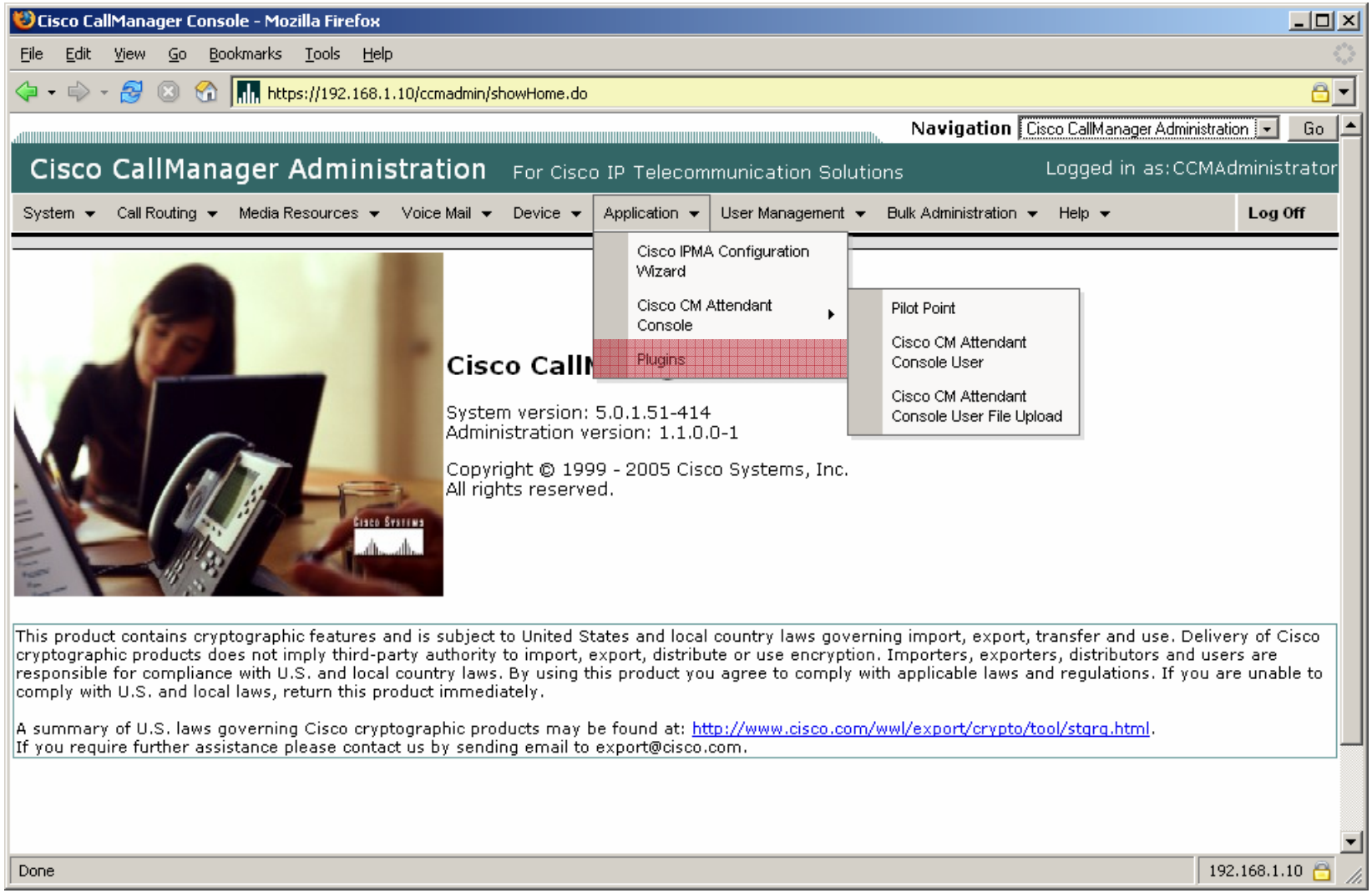
Cisco Unified CallManager 5.0 Application Menu: Plugins



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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.**

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Cisco Unified CallManager 5.0 User Management Menu: Users and Rights

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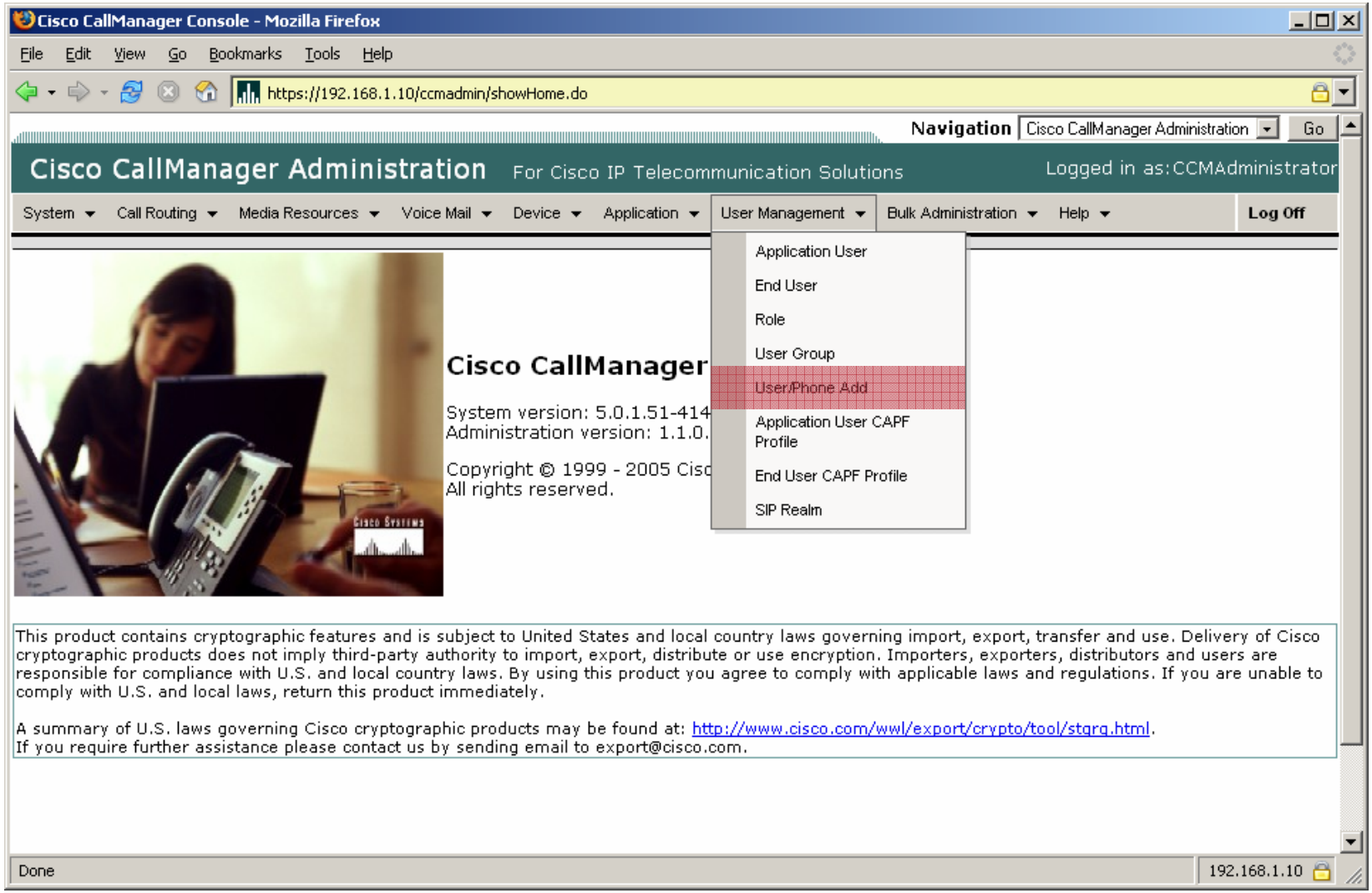
Cisco Unified CallManager 5.0 User Management Menu: Associate Users with Phones



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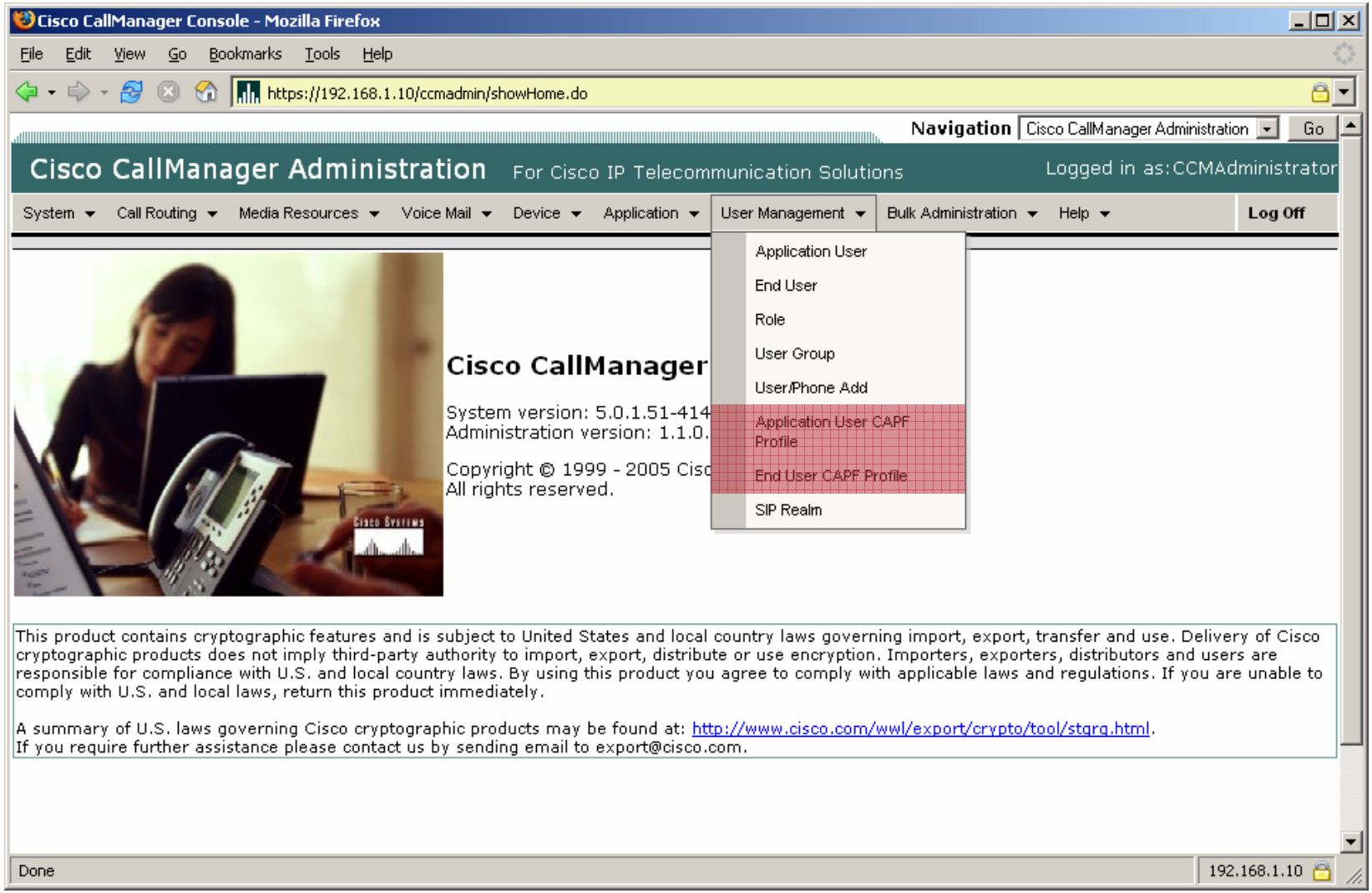
Cisco Unified CallManager 5.0 User Management Menu: CAPF



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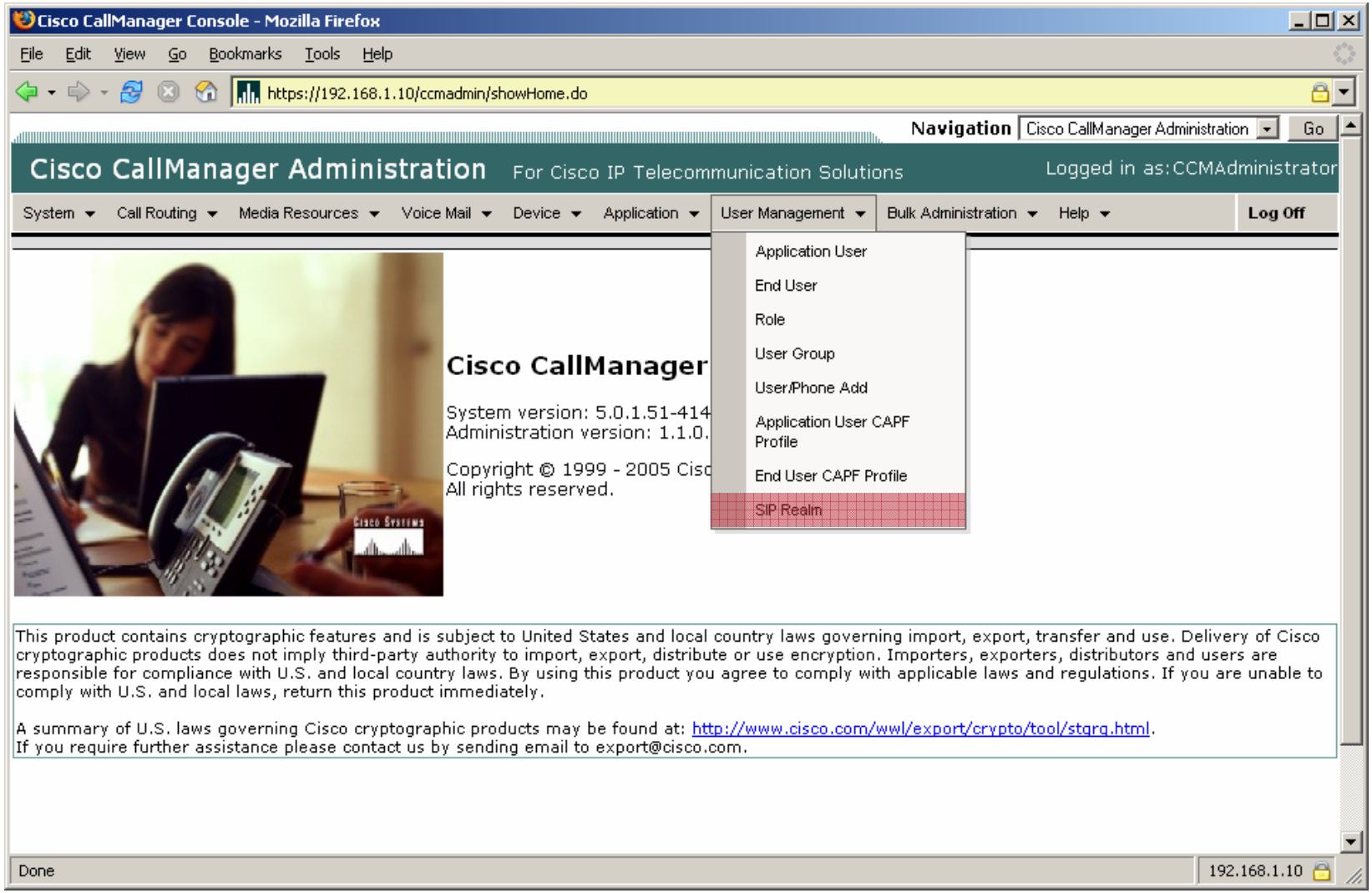
Cisco Unified CallManager 5.0 User Management Menu: SIP Realm



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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.**

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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)**

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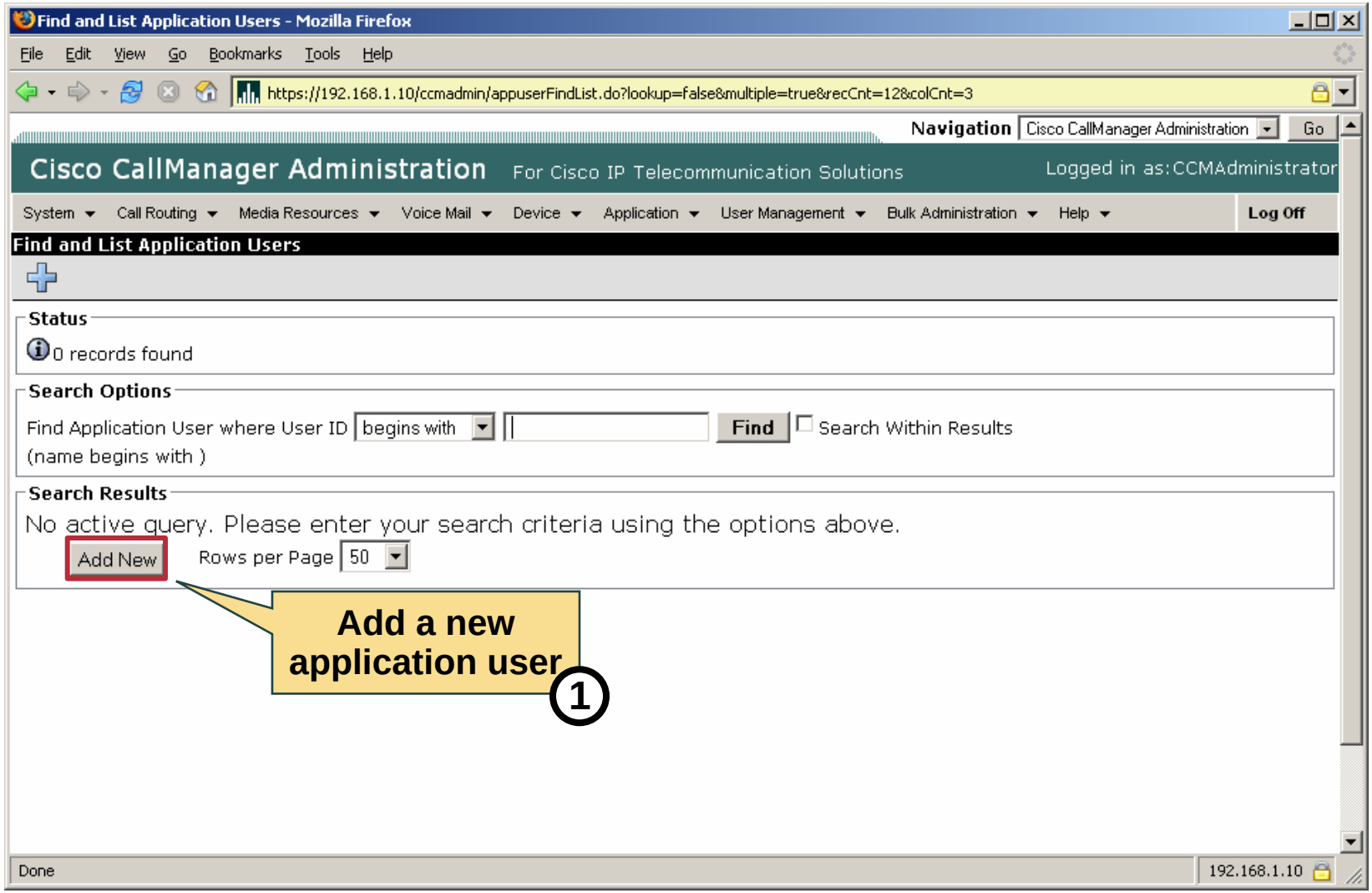
Multi Level Admin



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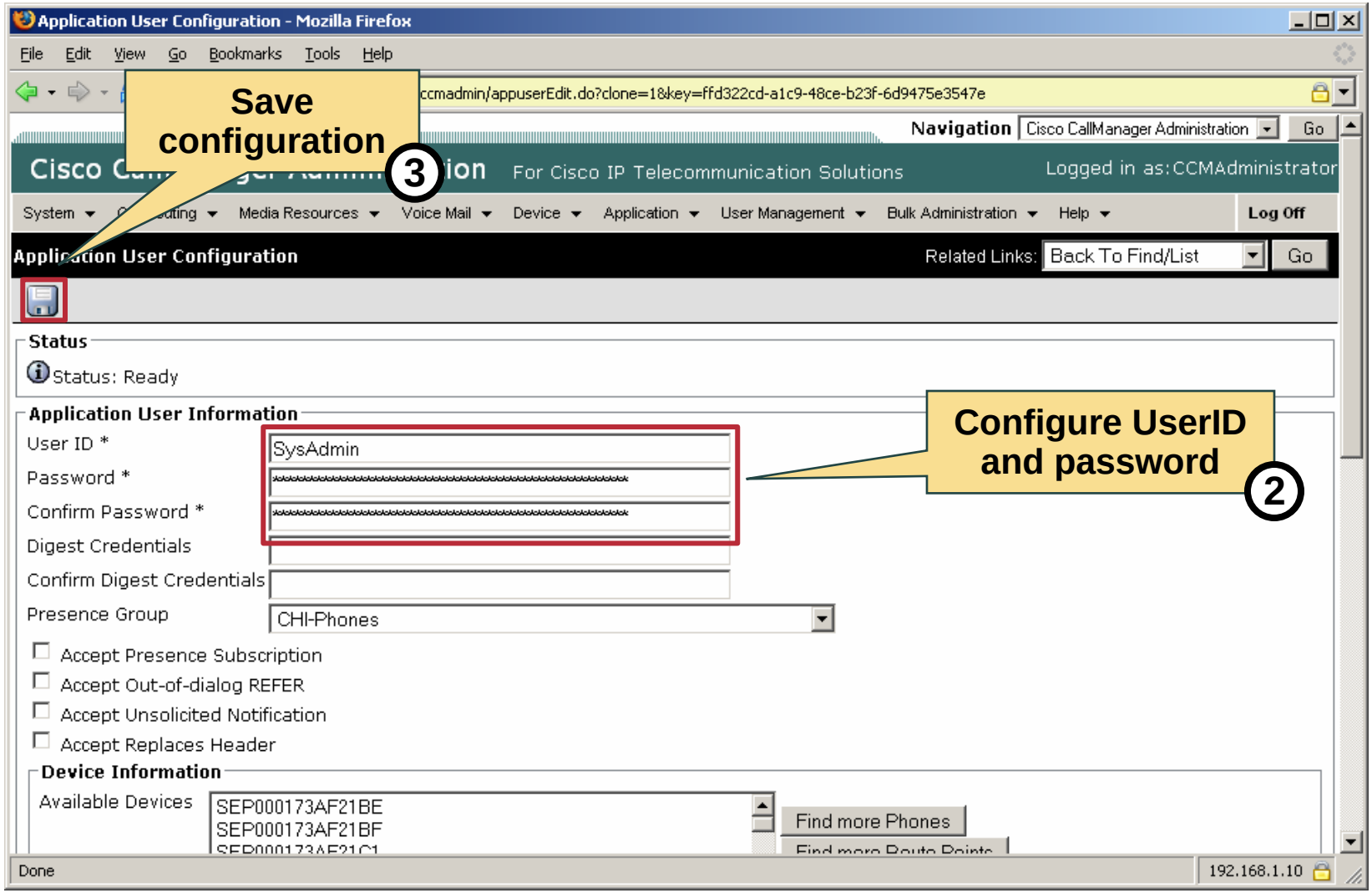
Multi Level Admin



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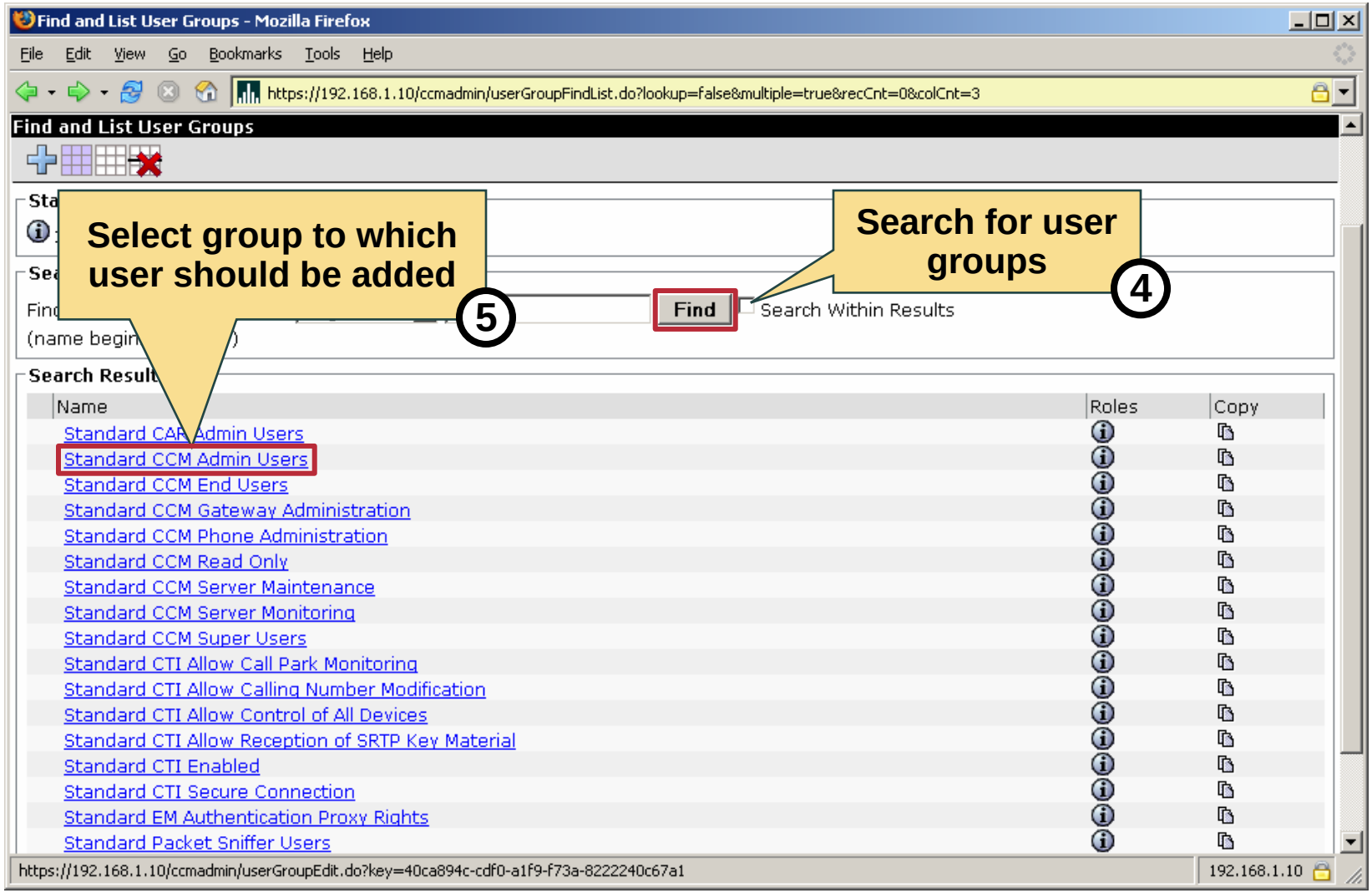
Multi Level Admin



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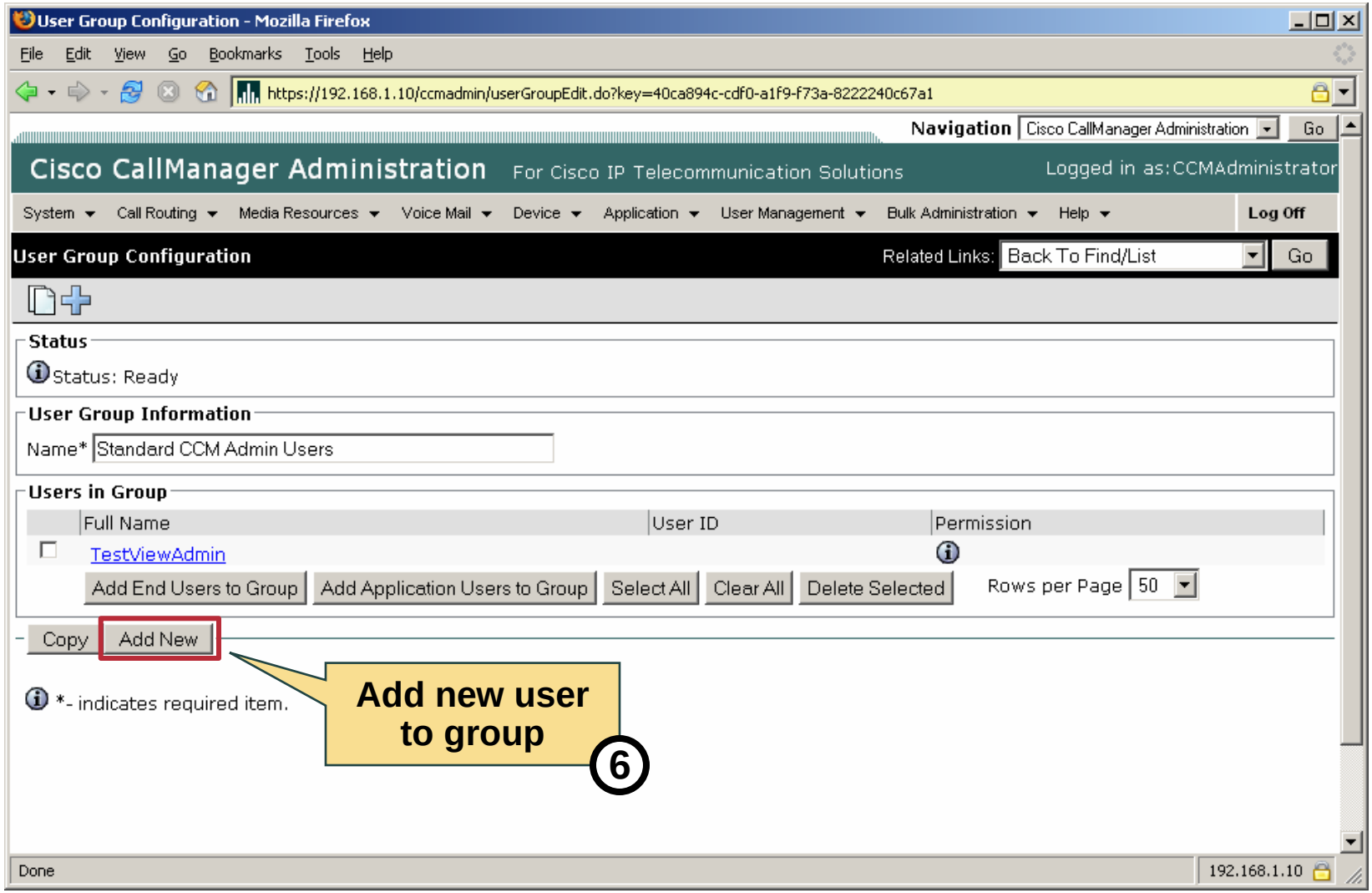
Multi Level Admin



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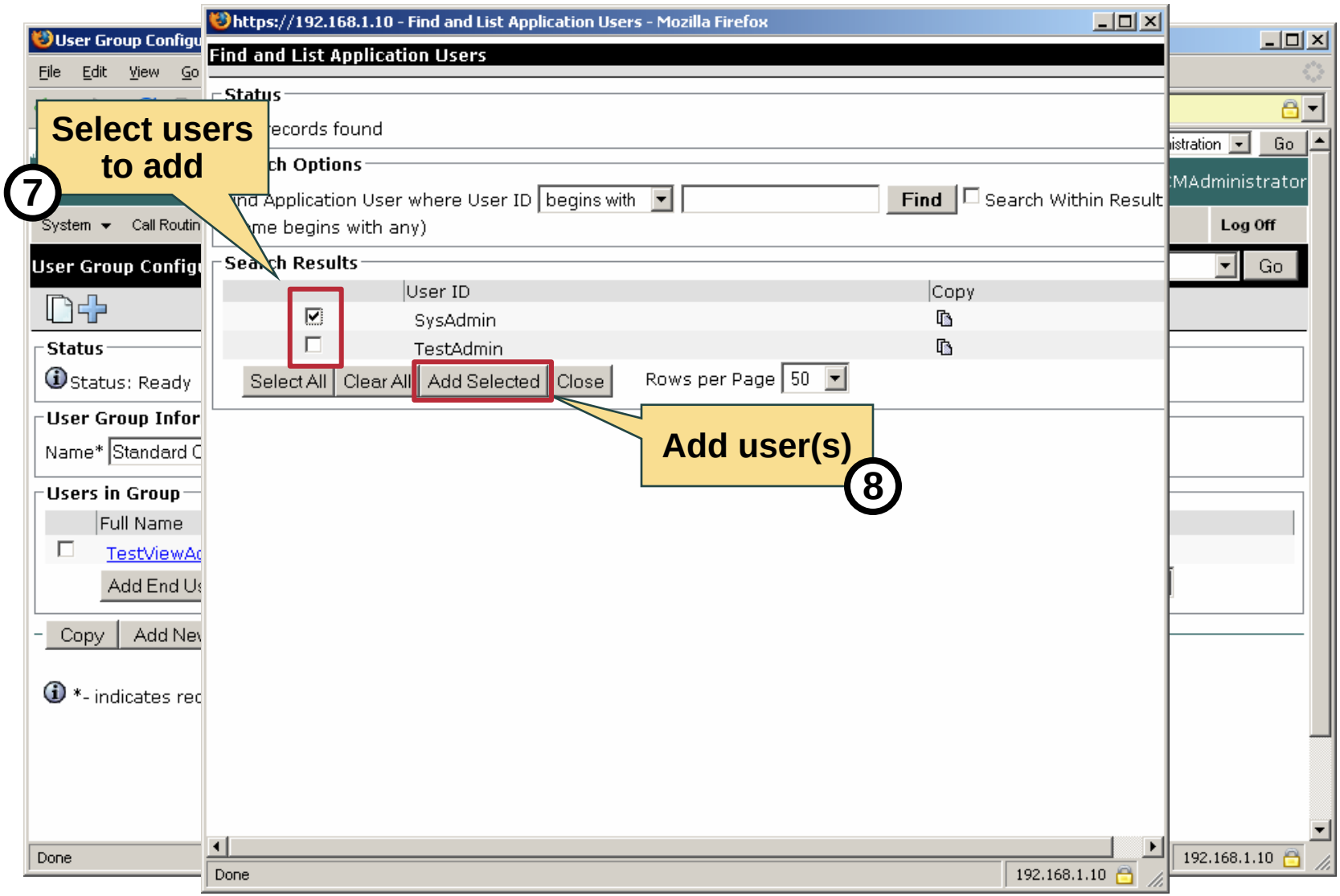
Multi Level Admin



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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.

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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.

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Administering Cisco Unified CallManager 5.0

Configuring CallManager 5.0 Basic Settings

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Objectives

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

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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.**

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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
• No need of DNS server. • Reduces potential source of failure. • Simplifies troubleshooting.	• Simplified management because of names instead of numbers. • Easier IP address changes because of name based IP paths.

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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.**

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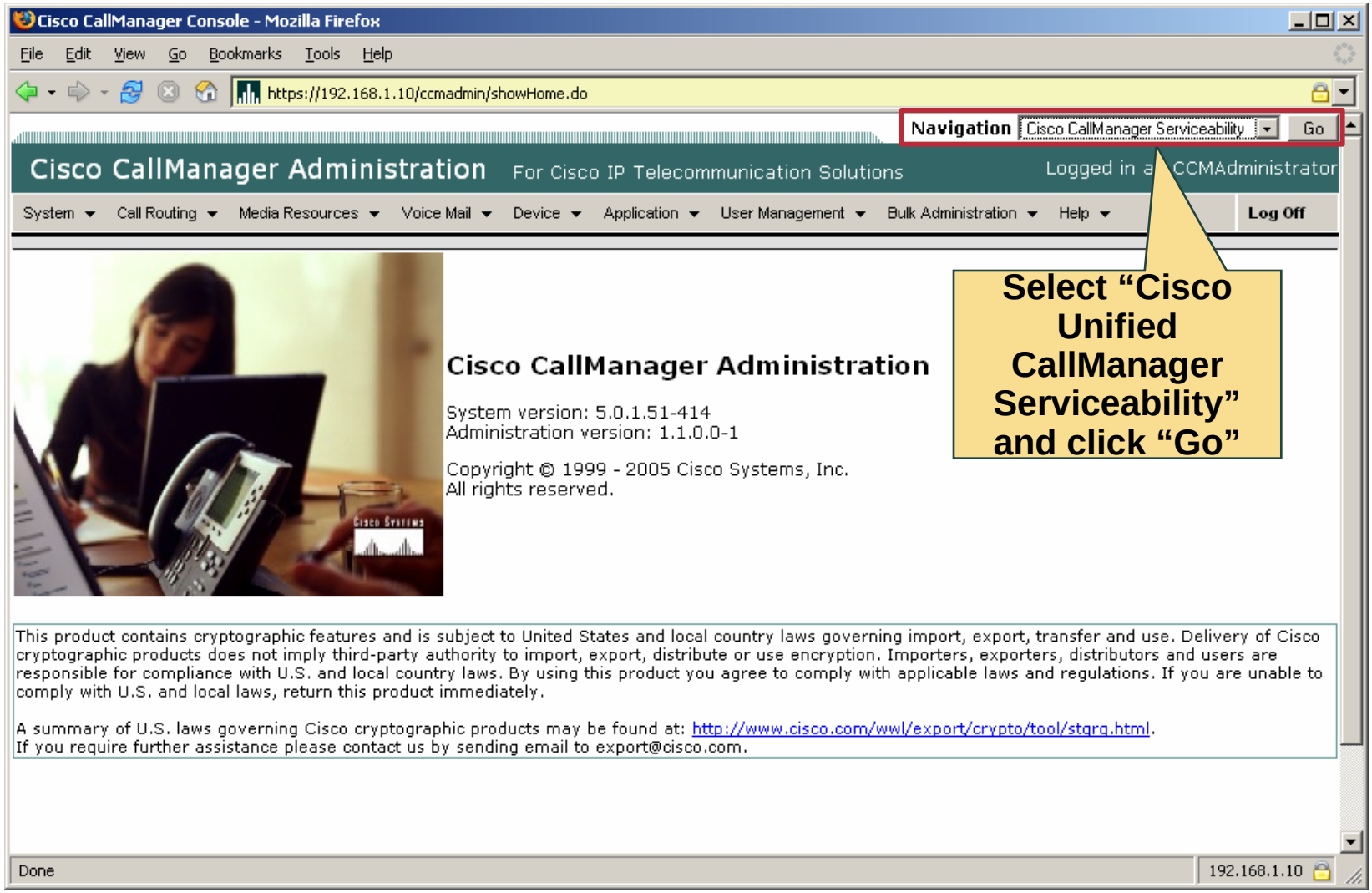
Service Activation



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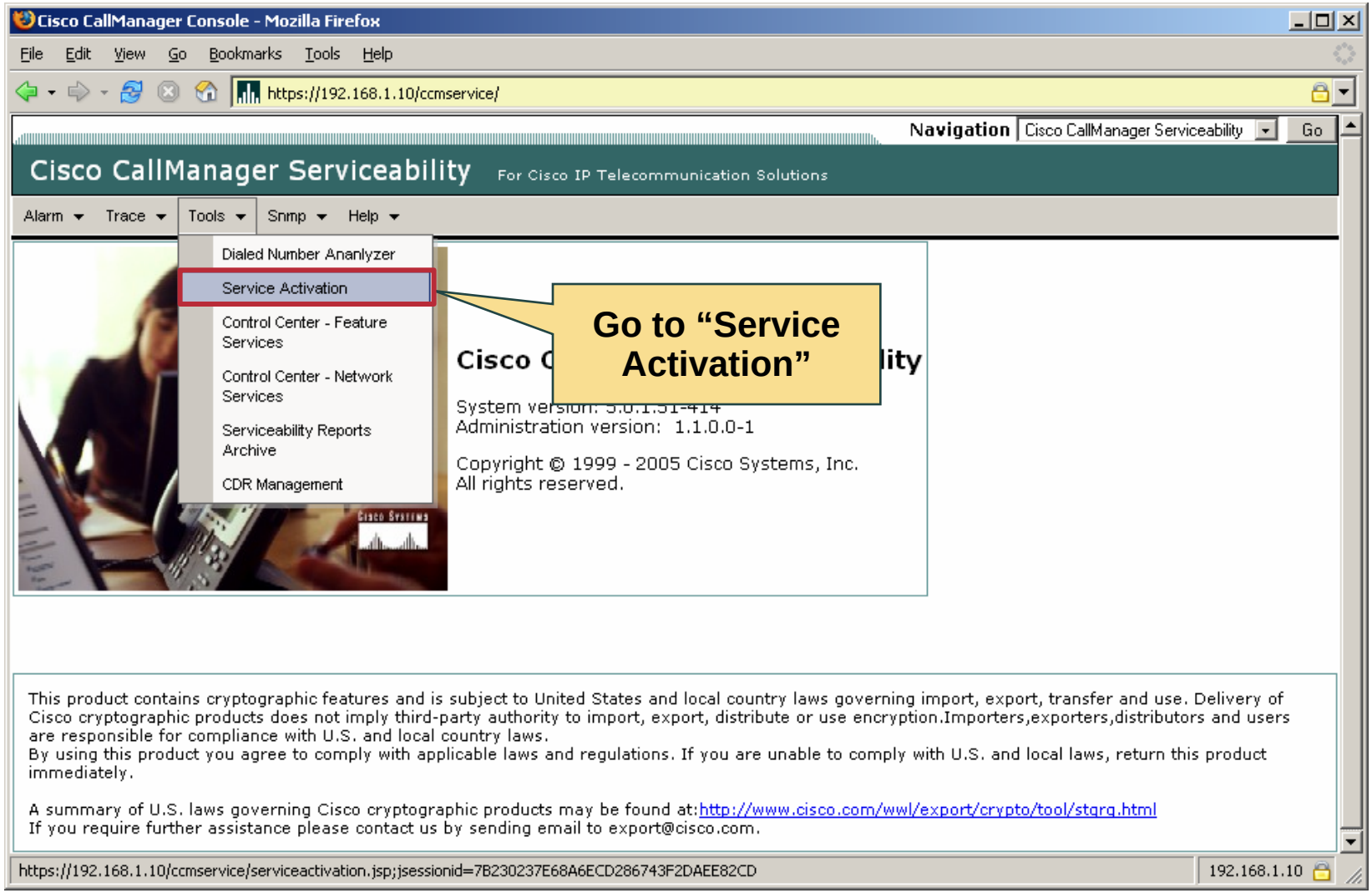
Service Activation



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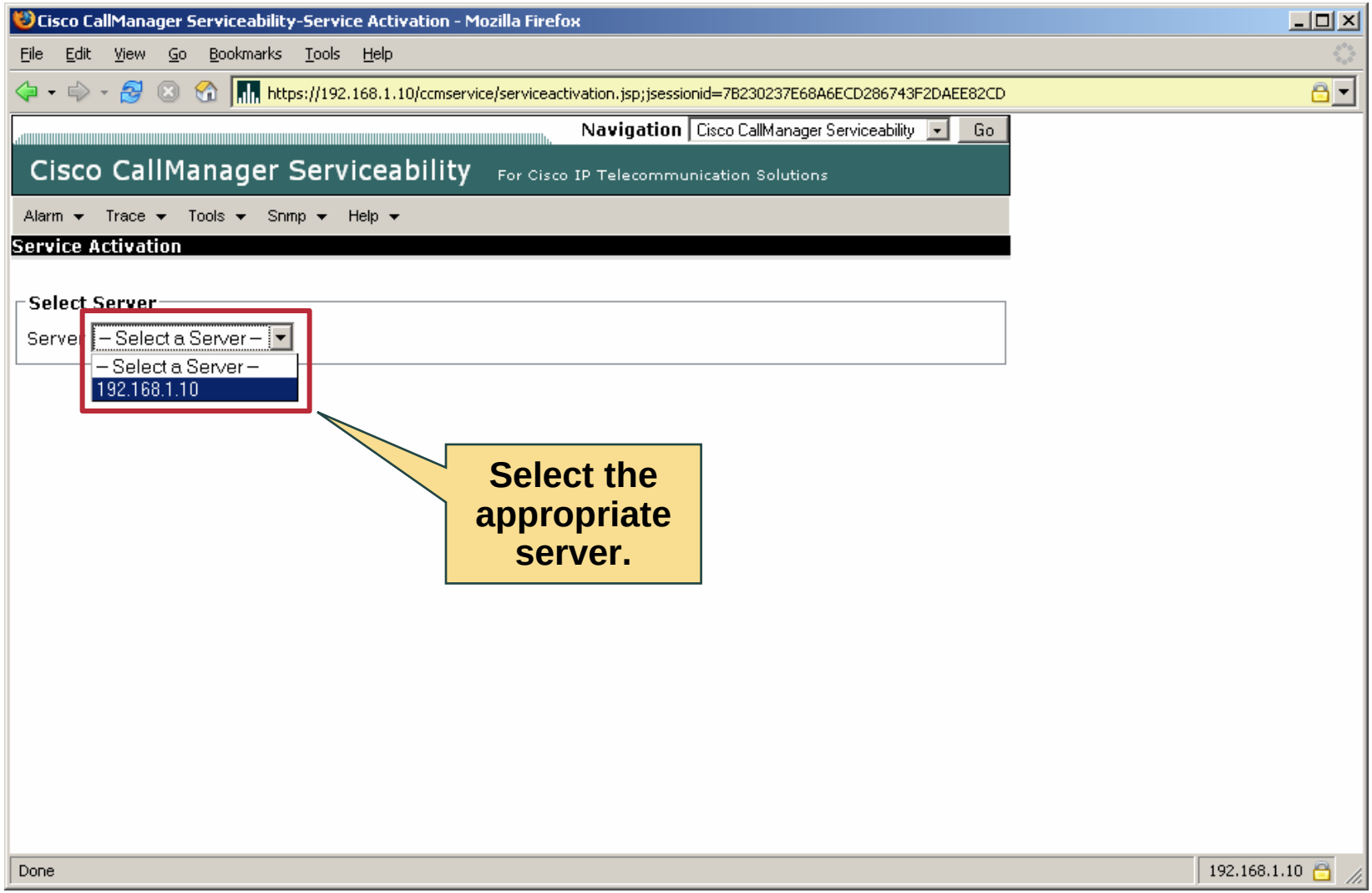
Service Activation



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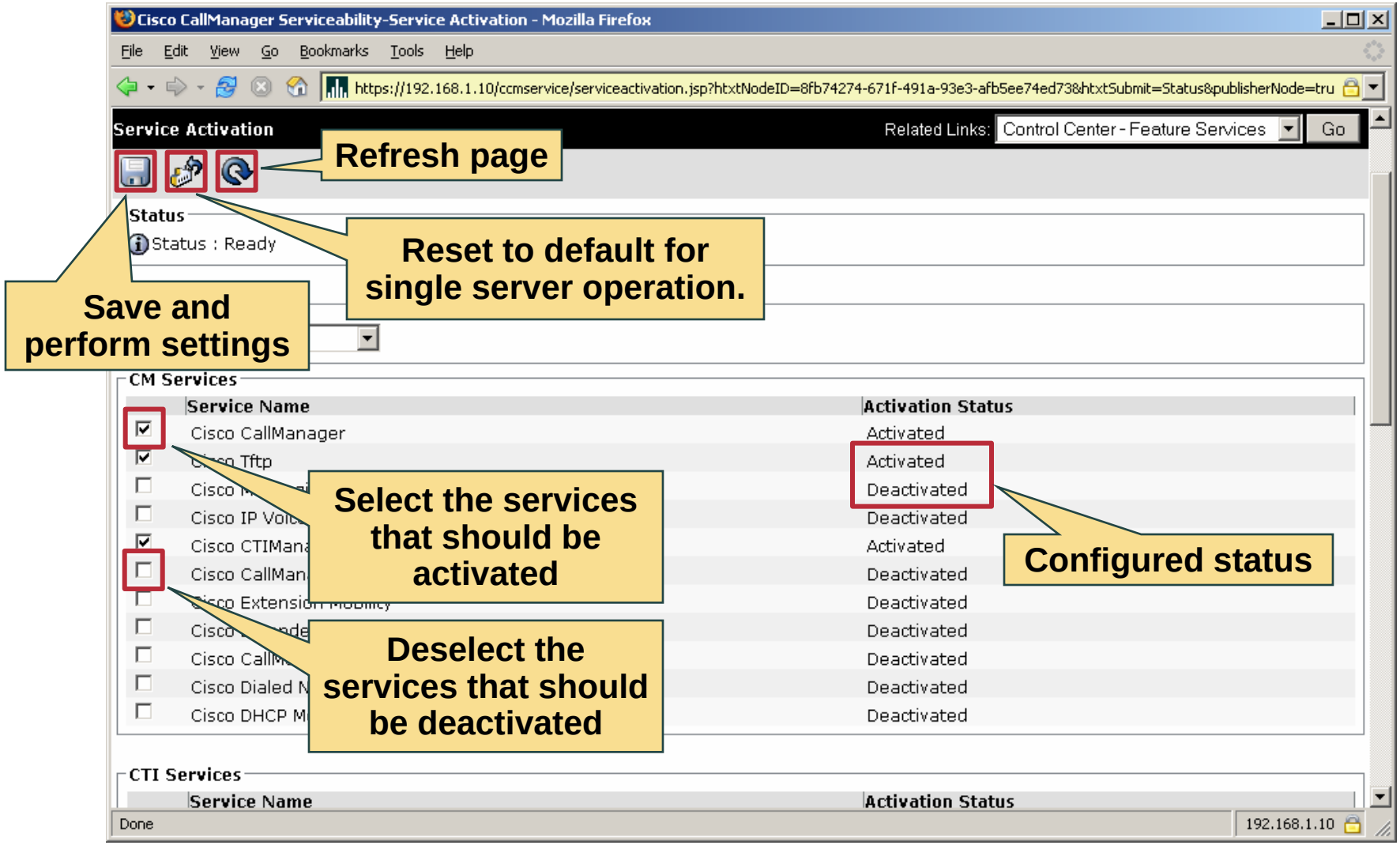
Service Activation



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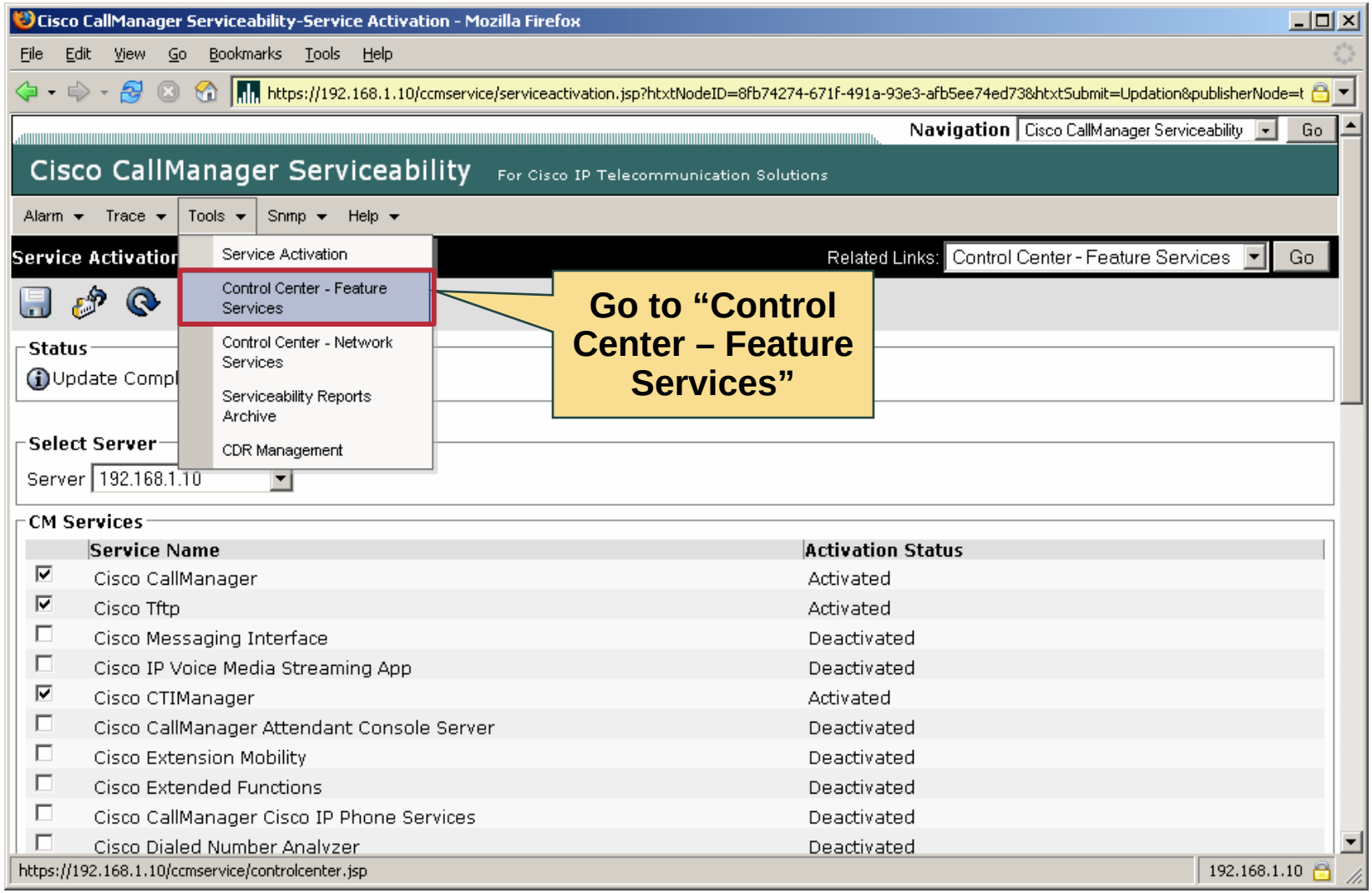
Control Center



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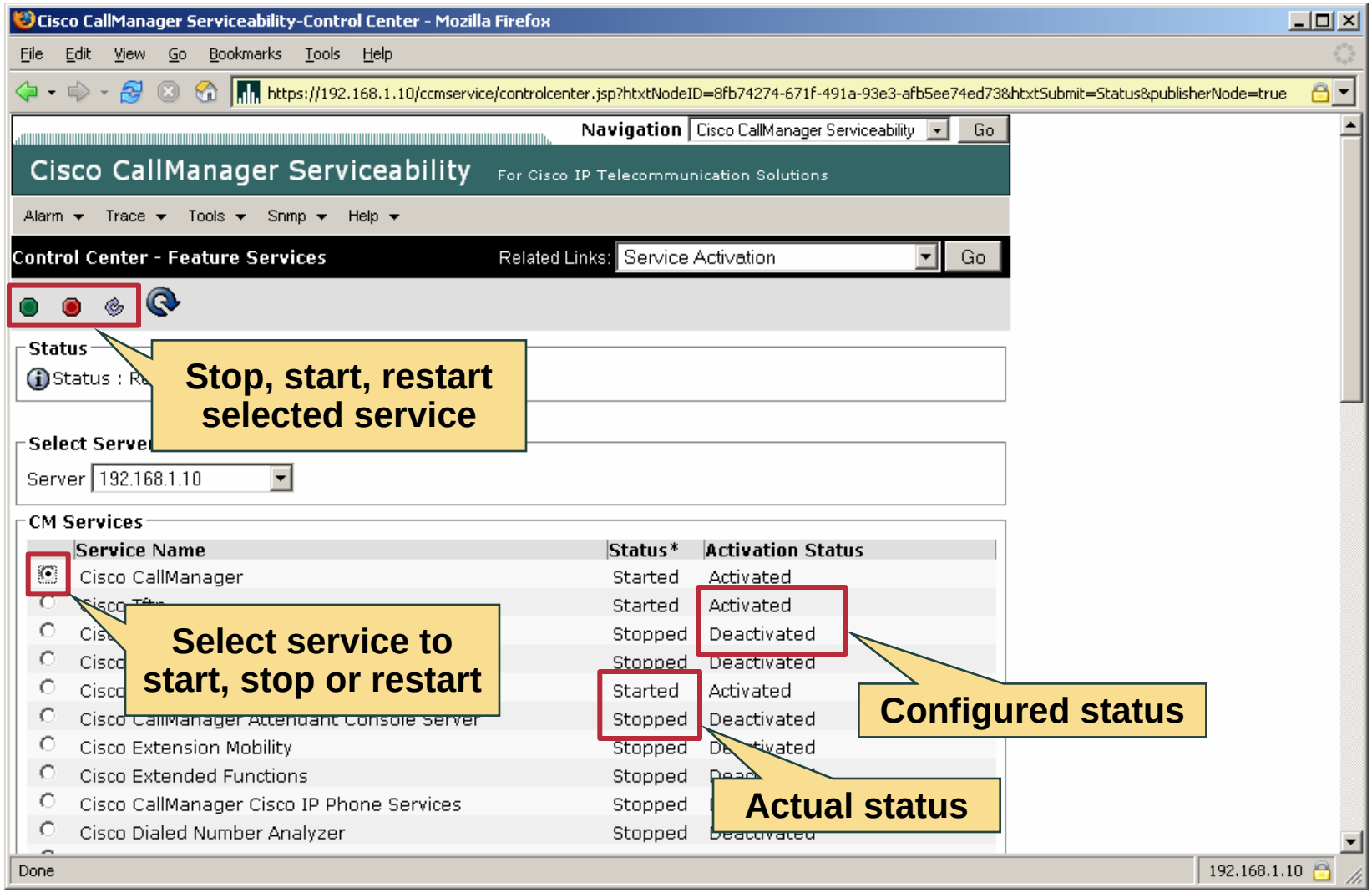
Control Center



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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.**

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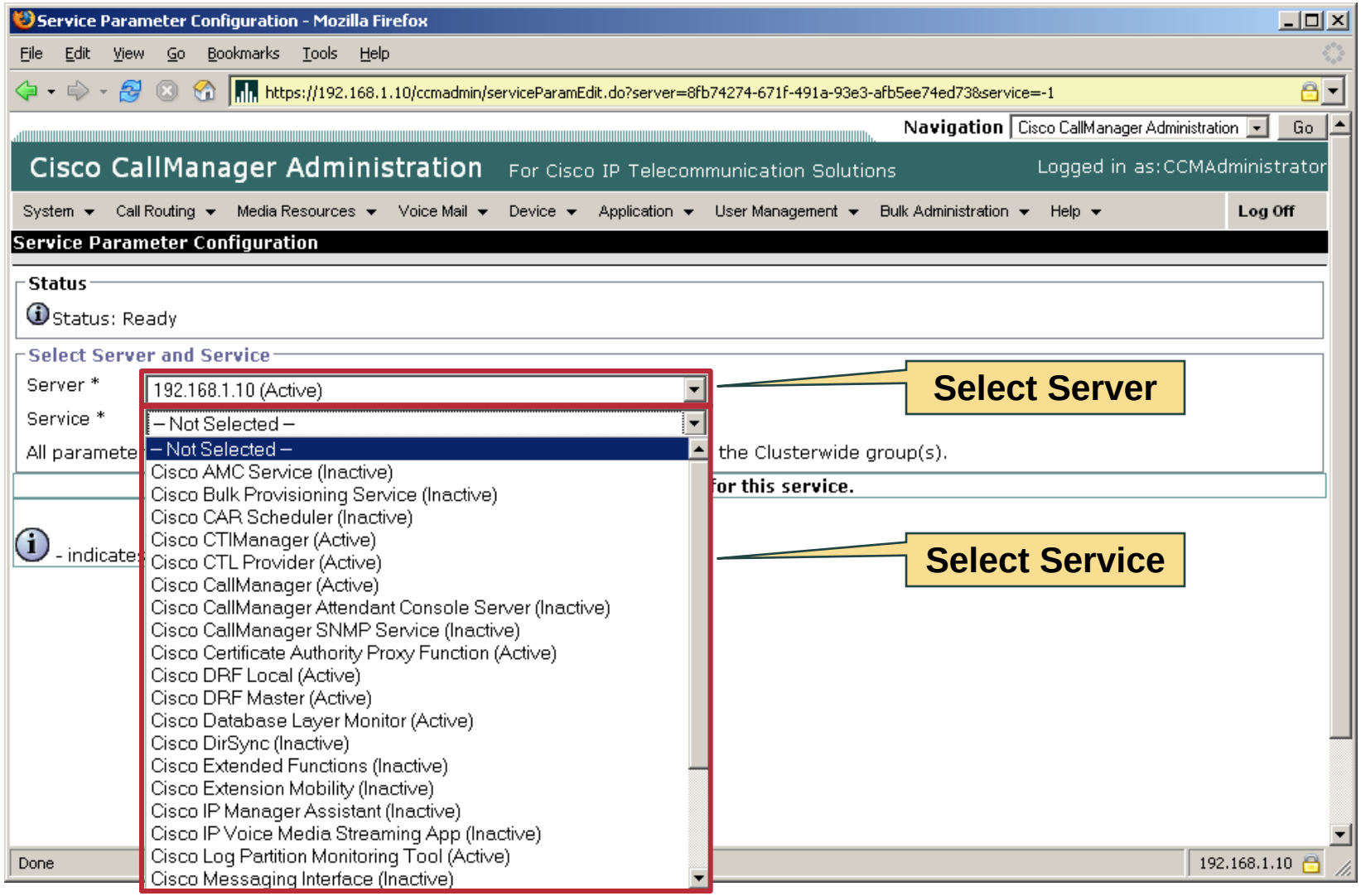
Service Parameters



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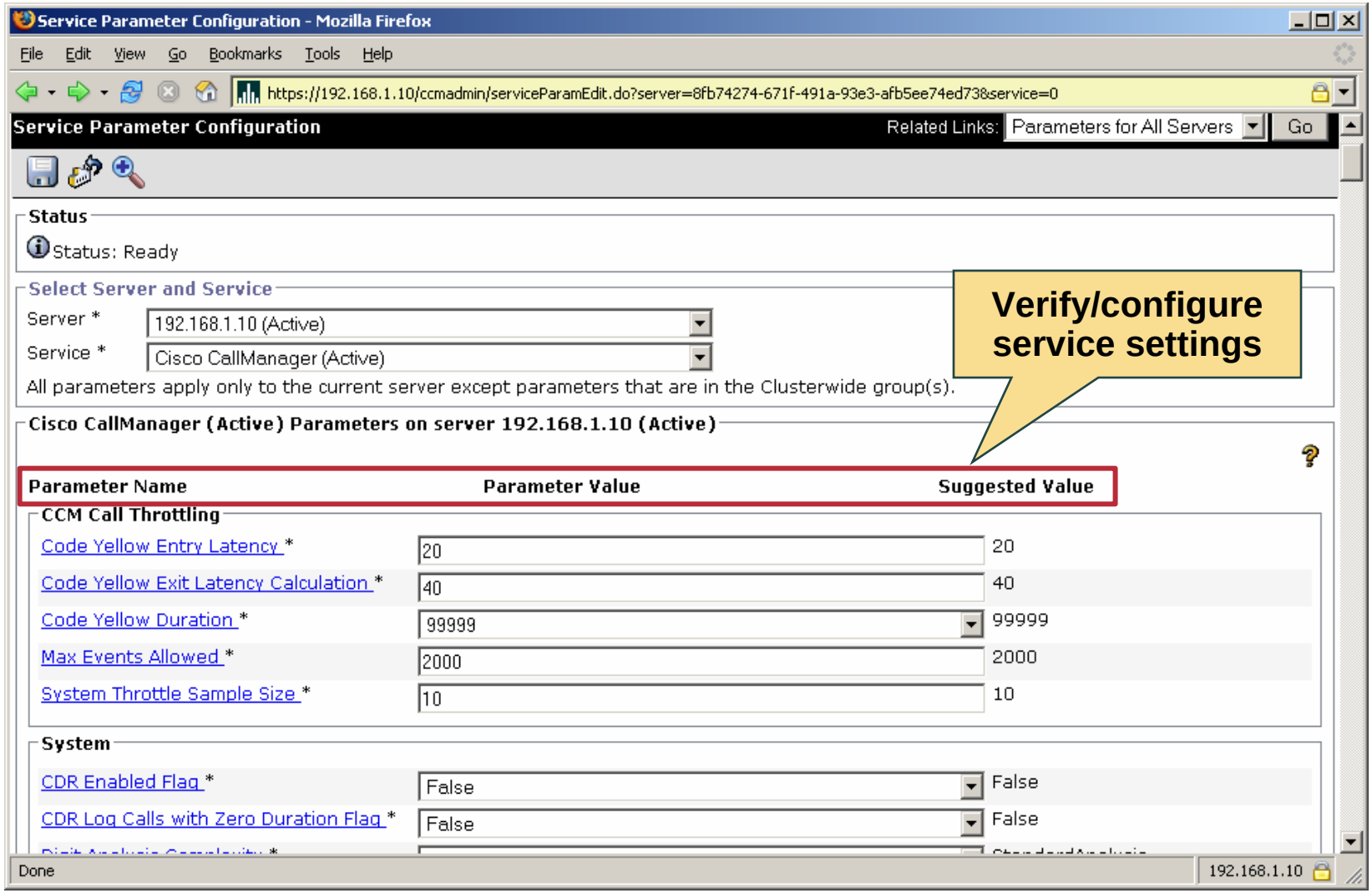
Service Parameters



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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.**

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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

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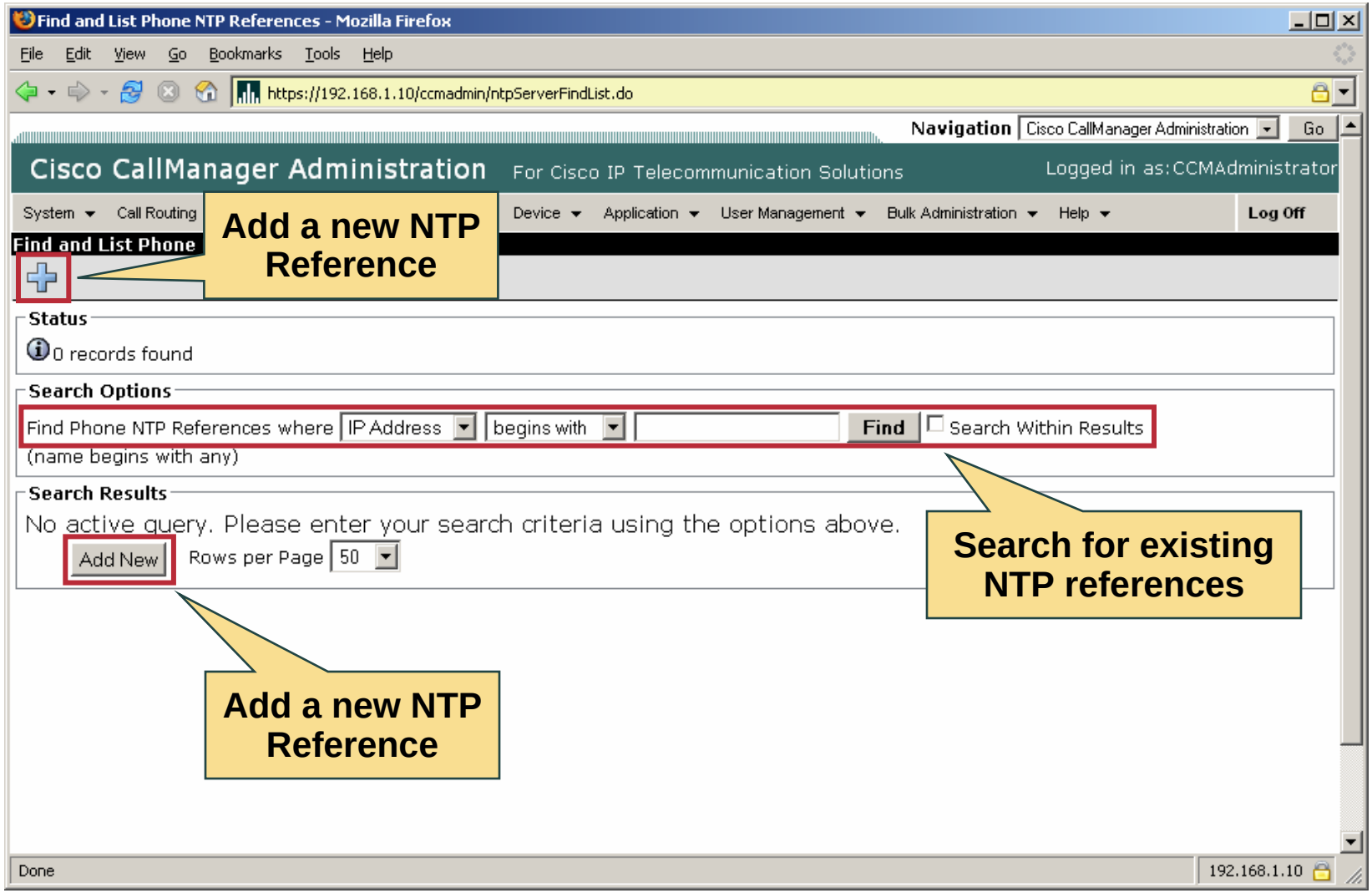
Phone NTP Reference



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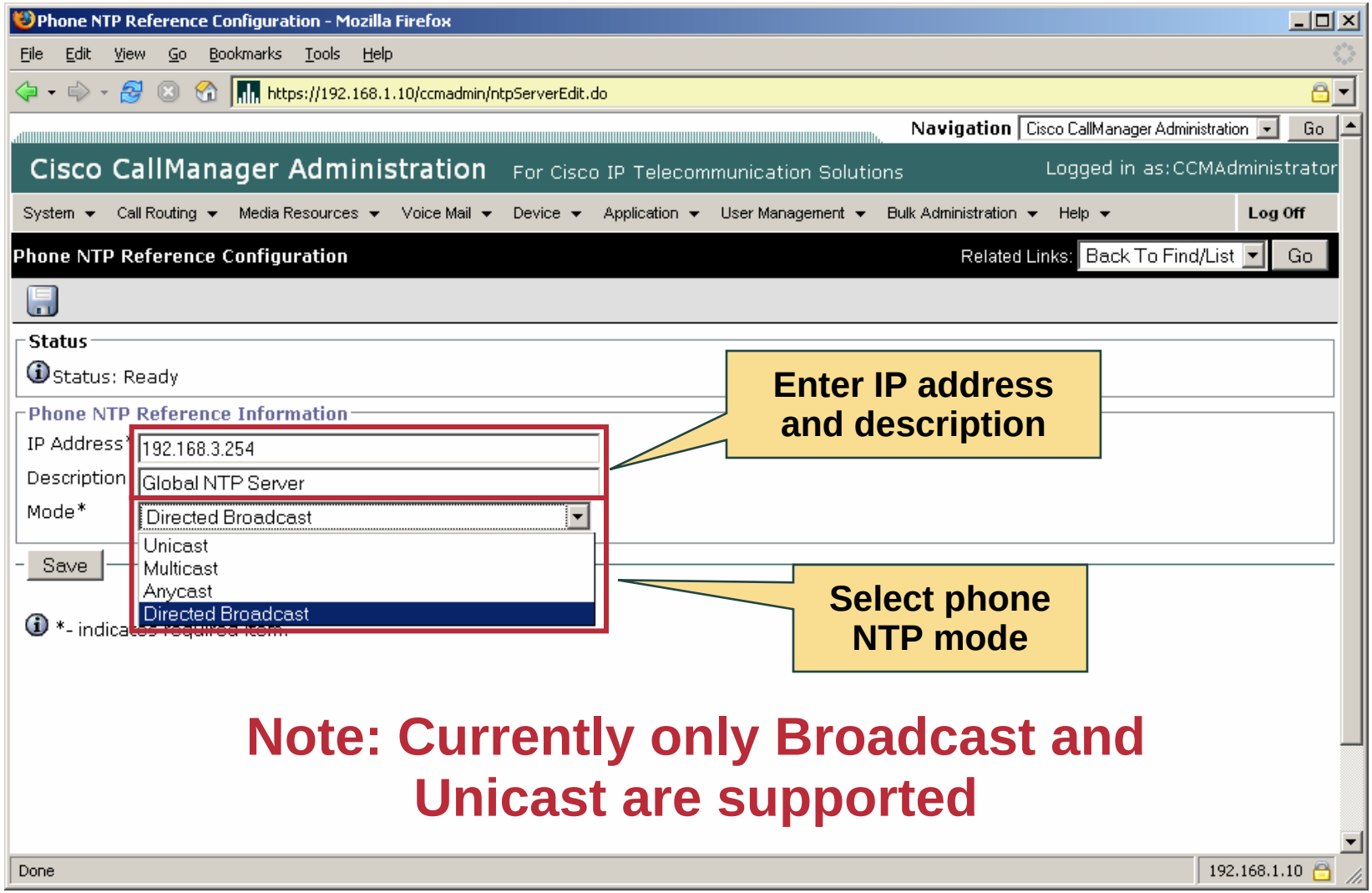
Phone NTP Reference



Note: Currently only Broadcast and Unicast are supported

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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.**

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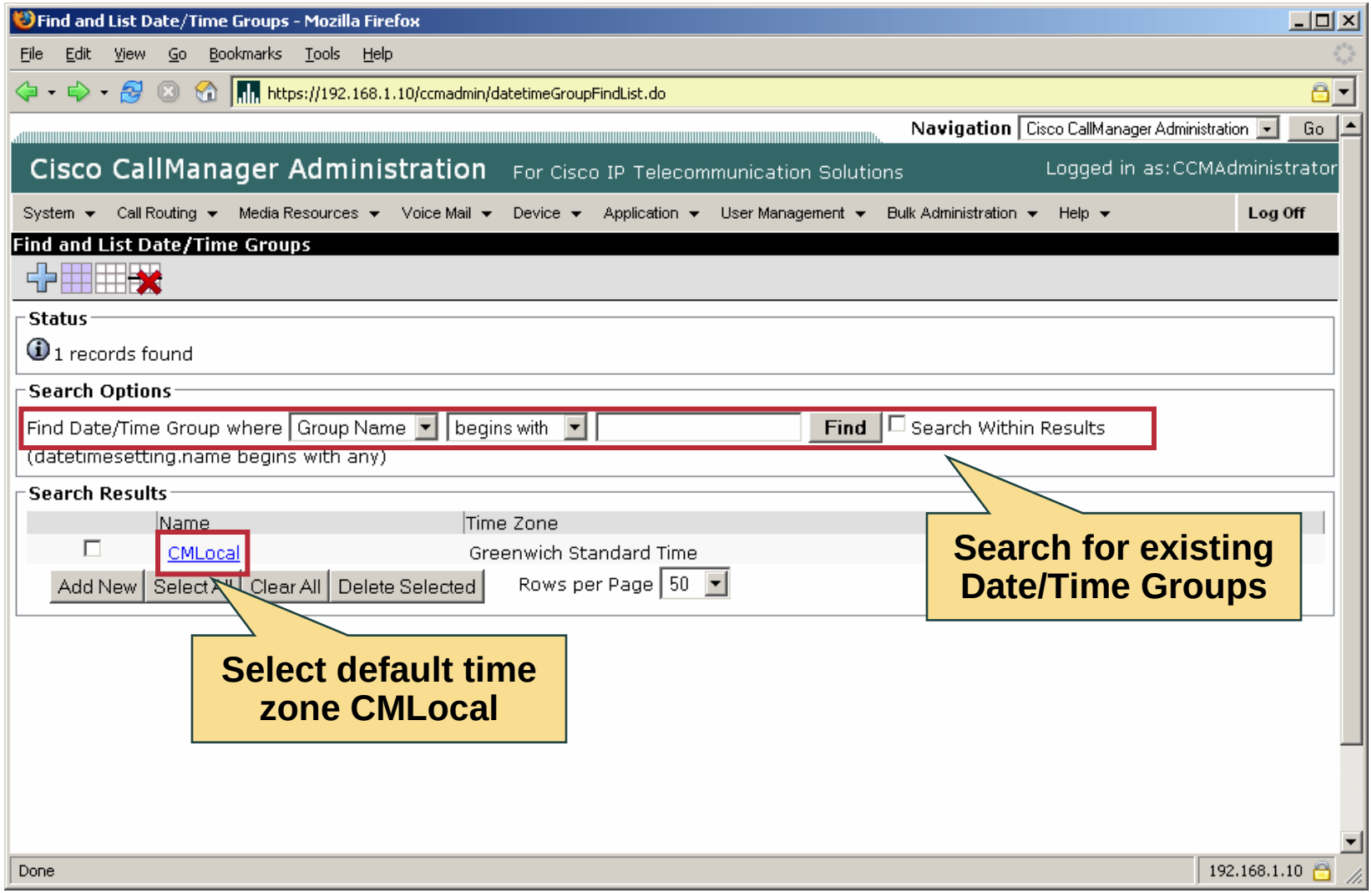
Time Zone



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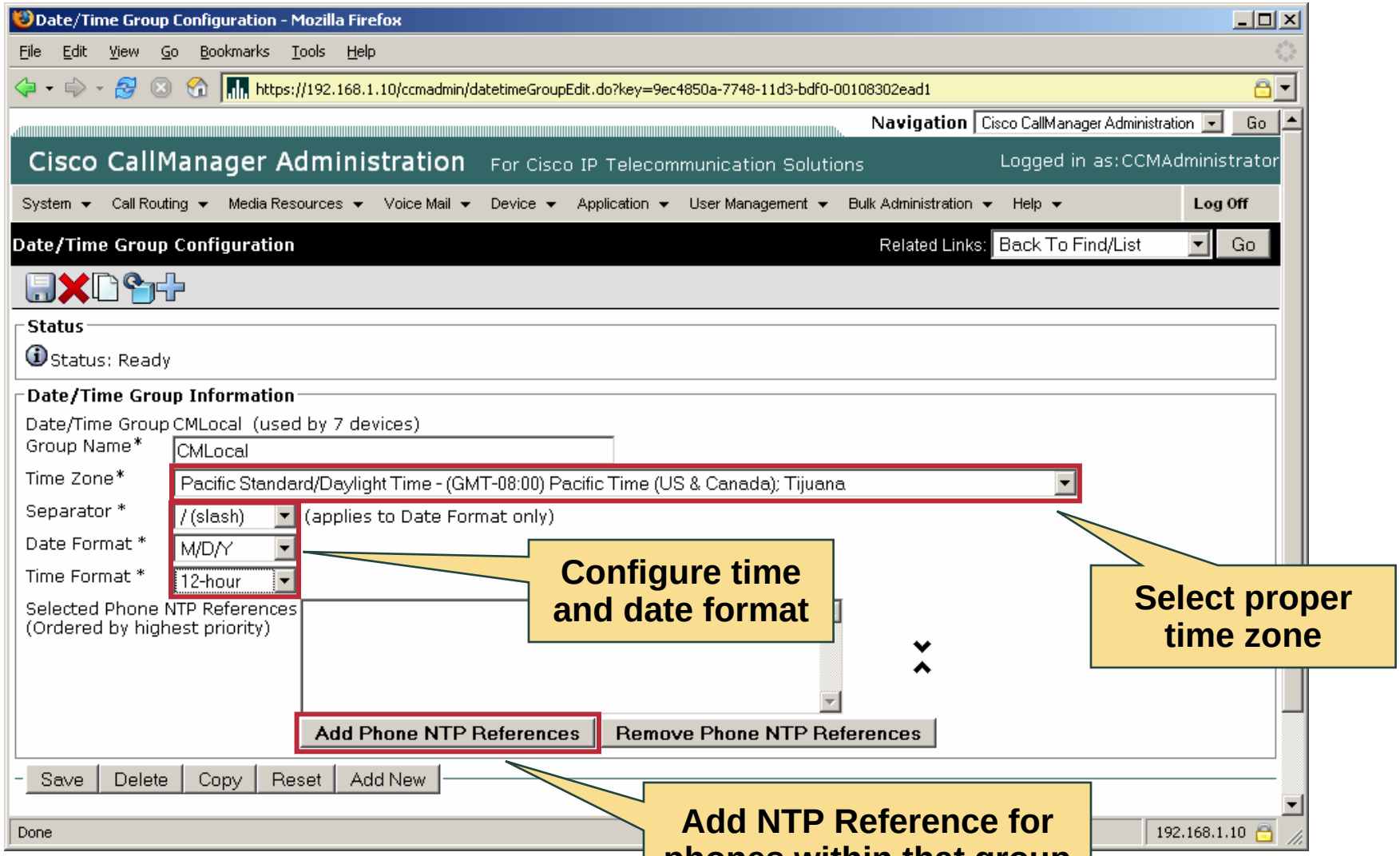
Time Zone



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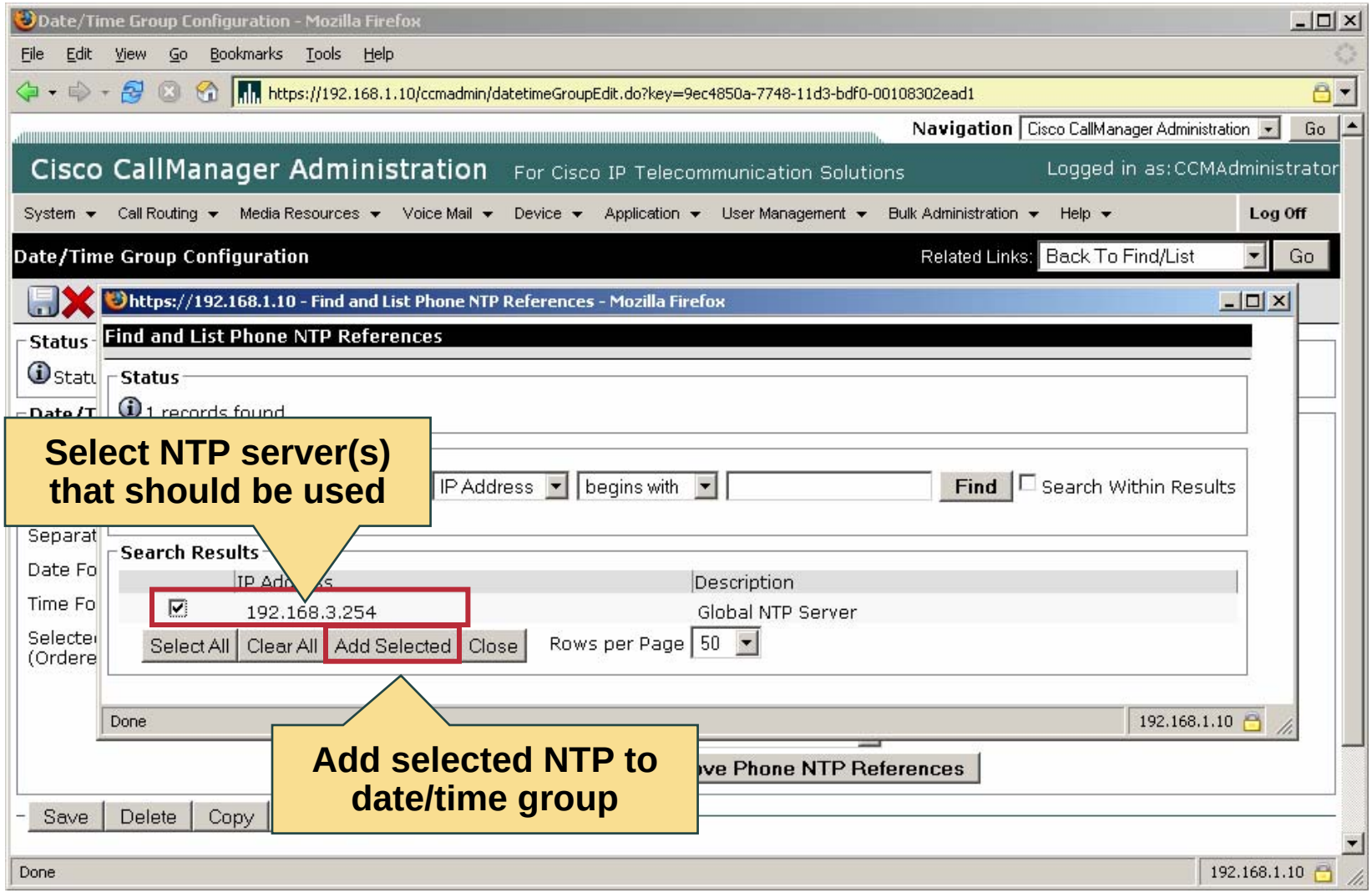
Time Zone



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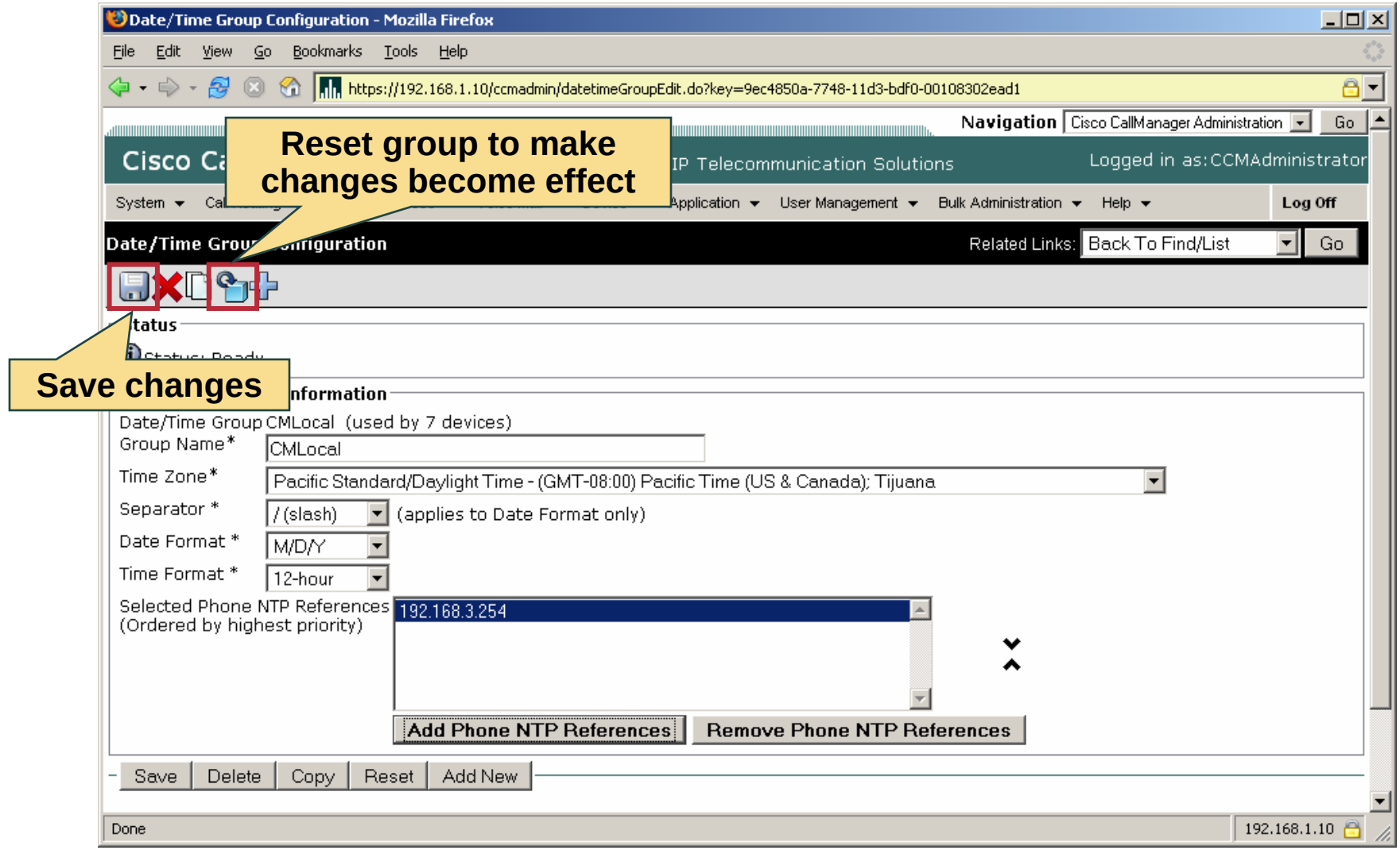
Time Zone



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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.**

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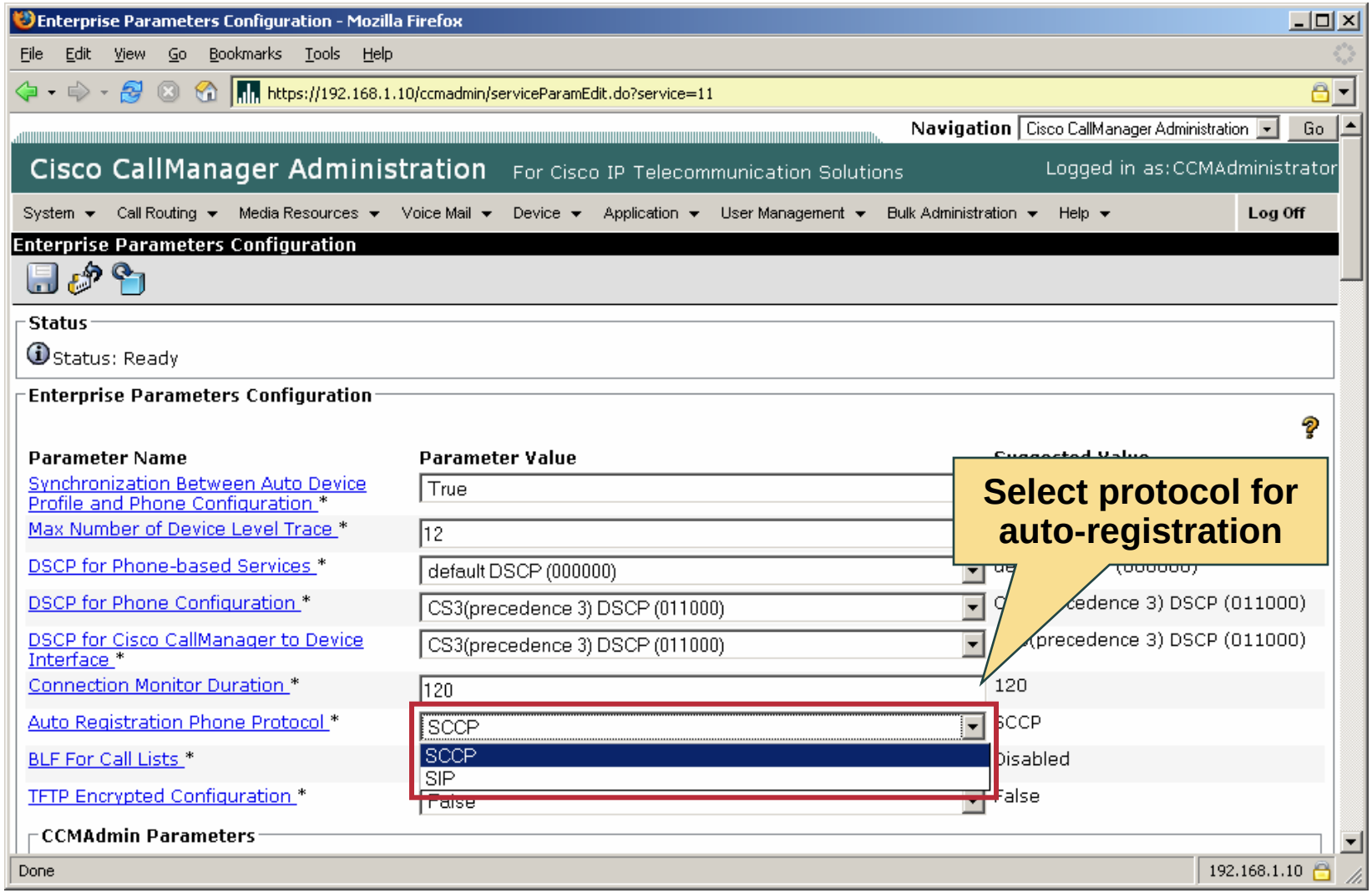
Enterprise Parameters



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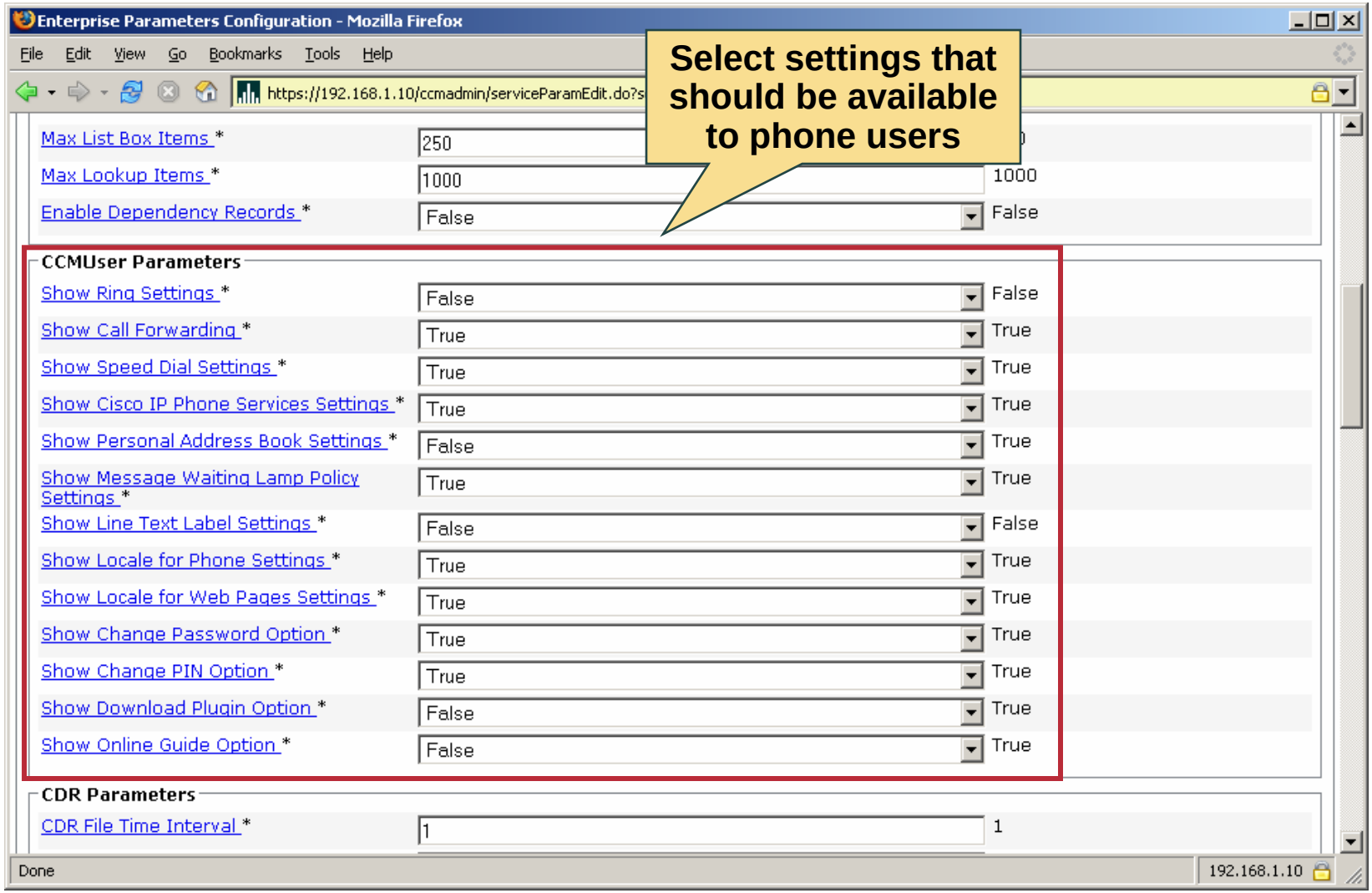
Enterprise Parameters



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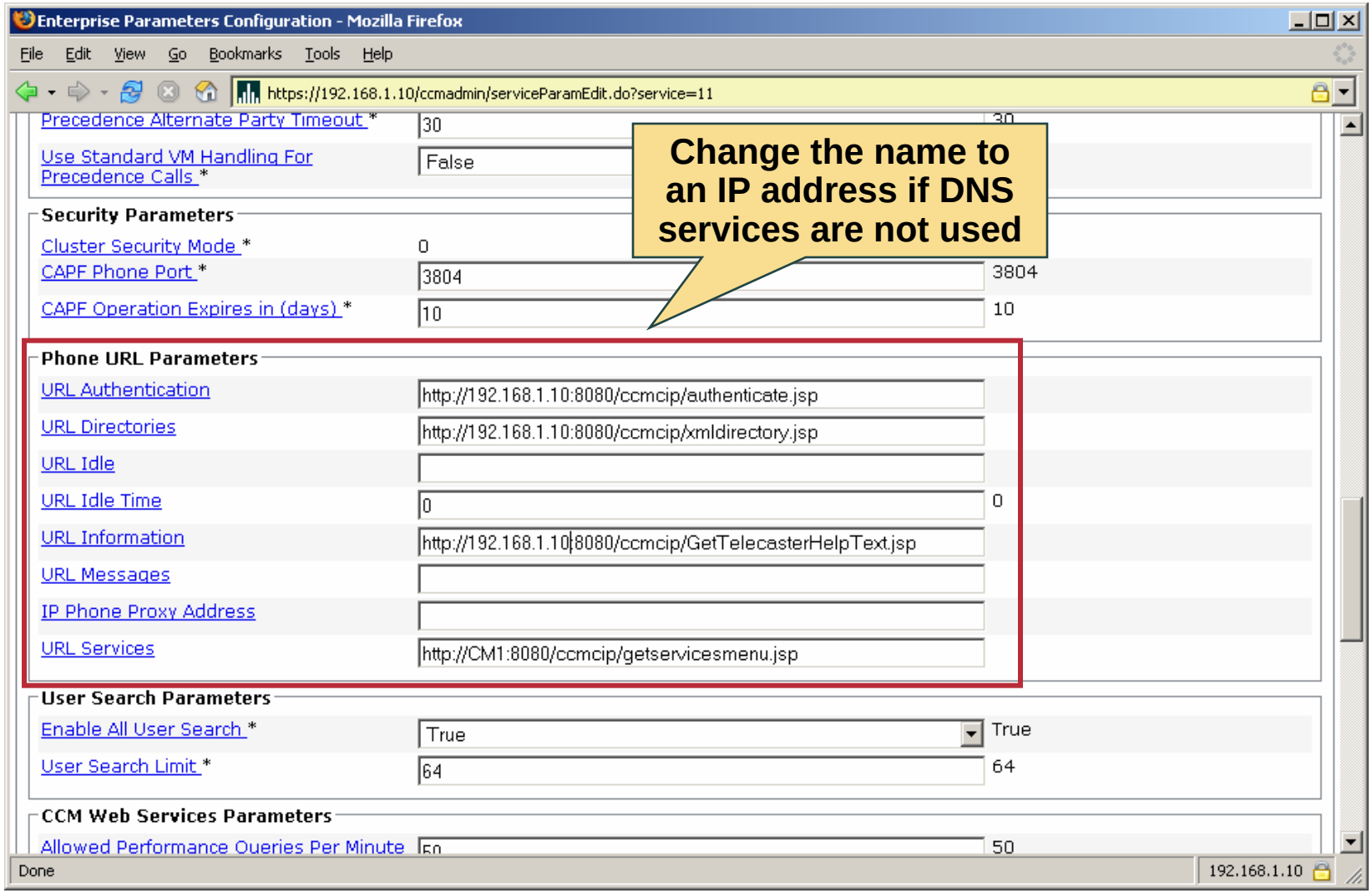
Enterprise Parameters



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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.

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Administering Cisco Unified CallManager 5.0

Serviceability and Platform Administration

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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

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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.**

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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.

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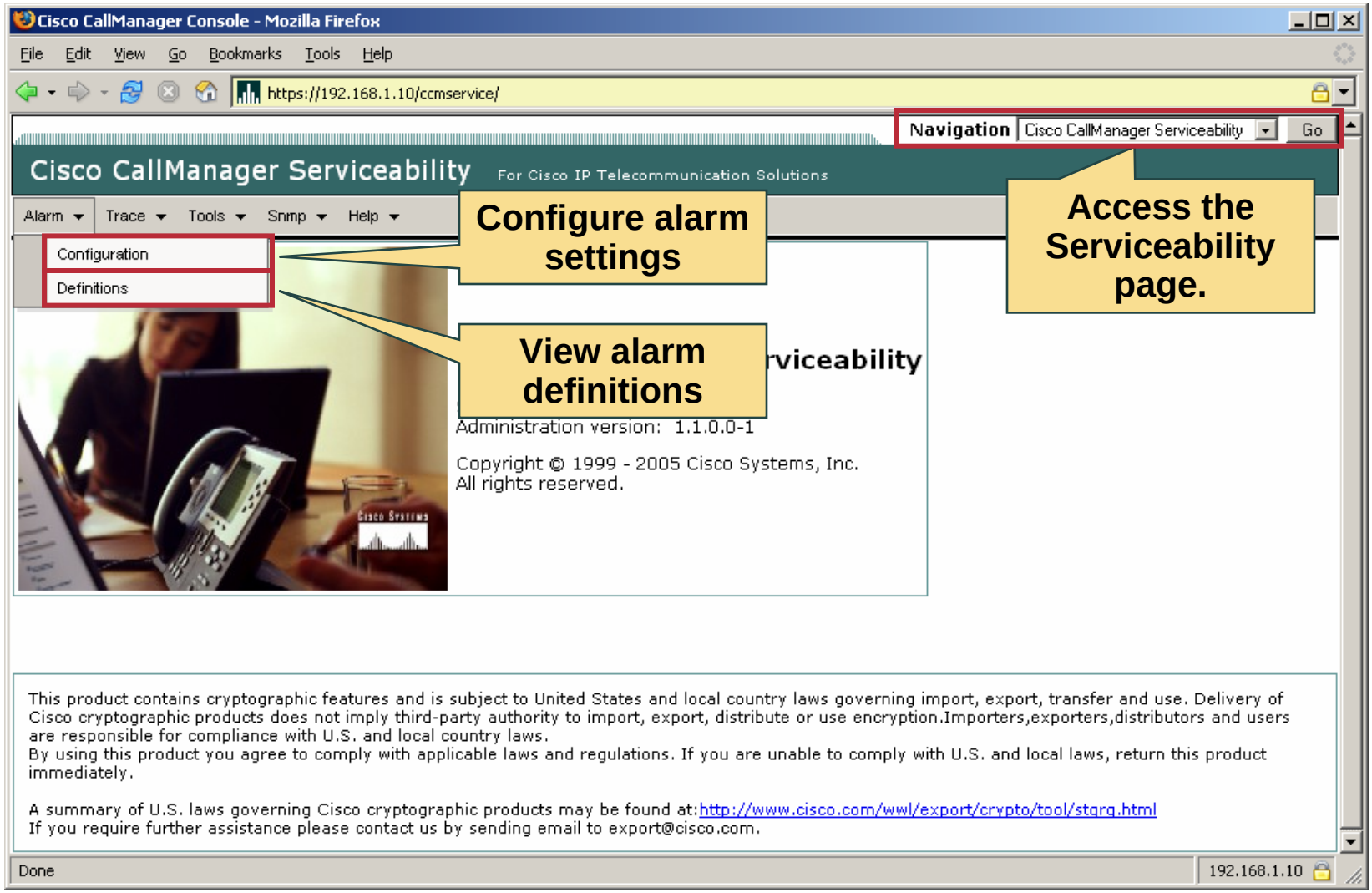
Cisco Unified CallManager 5.0 Alarms



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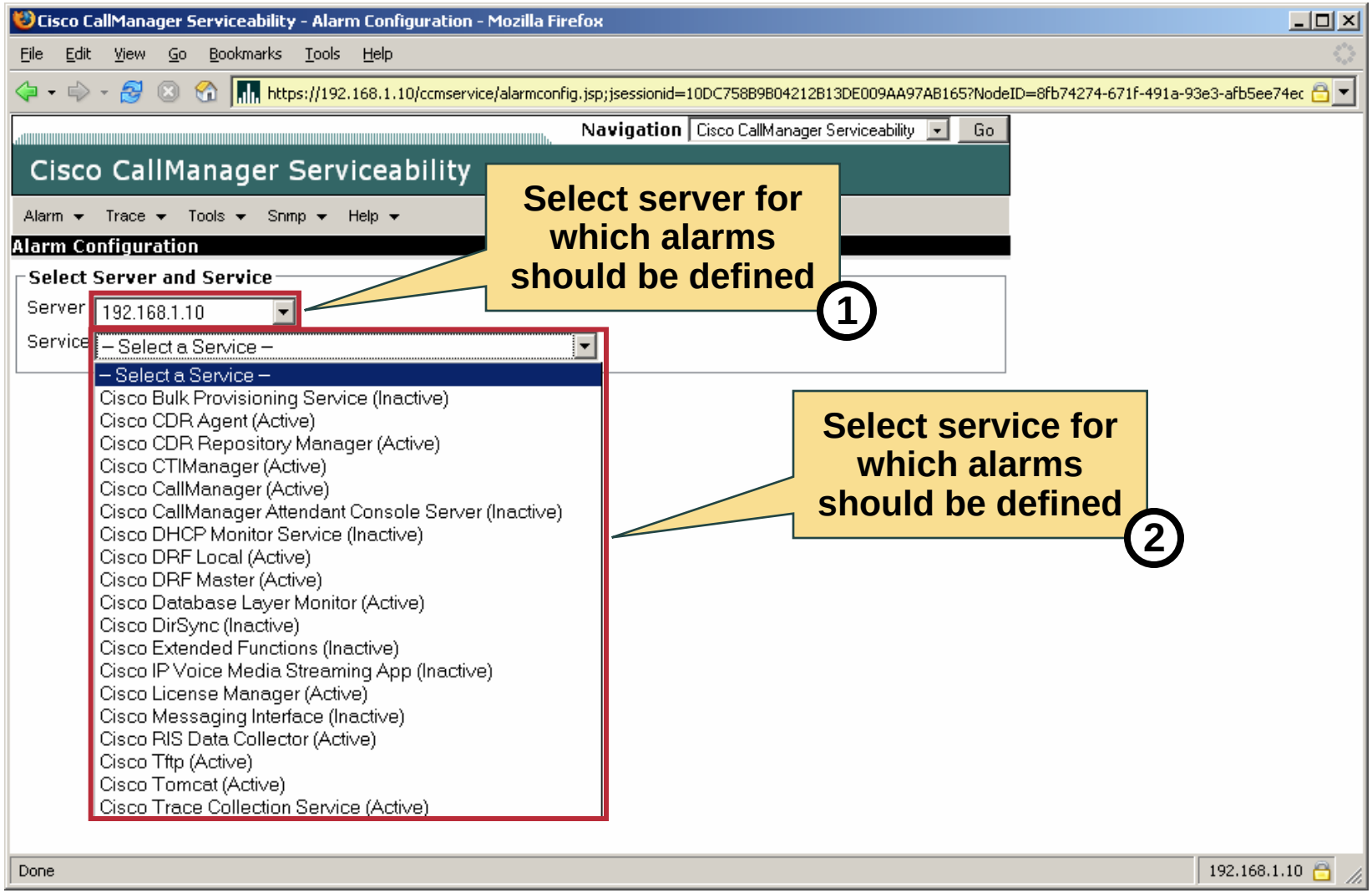
Cisco Unified CallManager 5.0 Alarms



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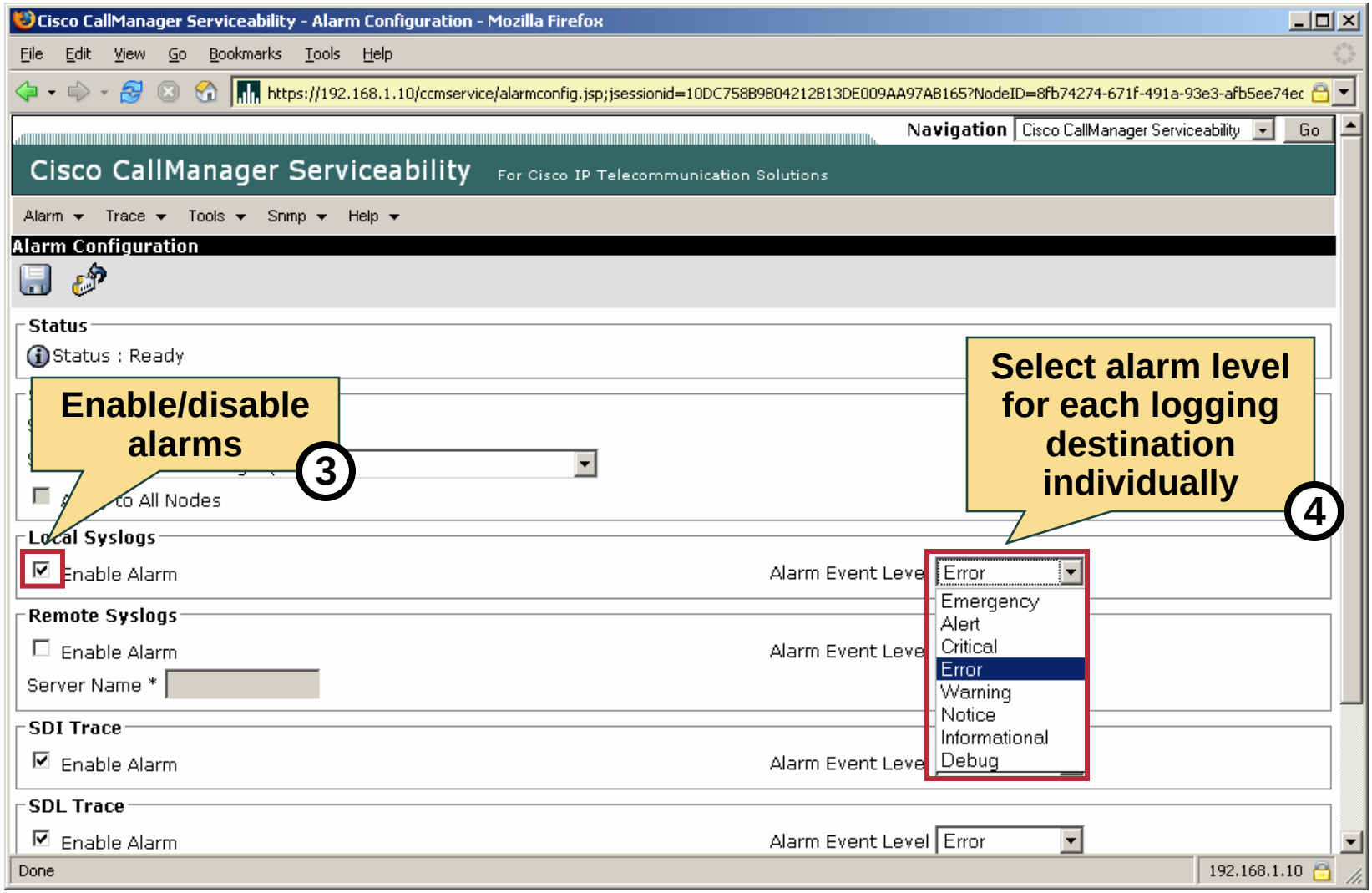
Cisco Unified CallManager 5.0 Alarms



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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.

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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.

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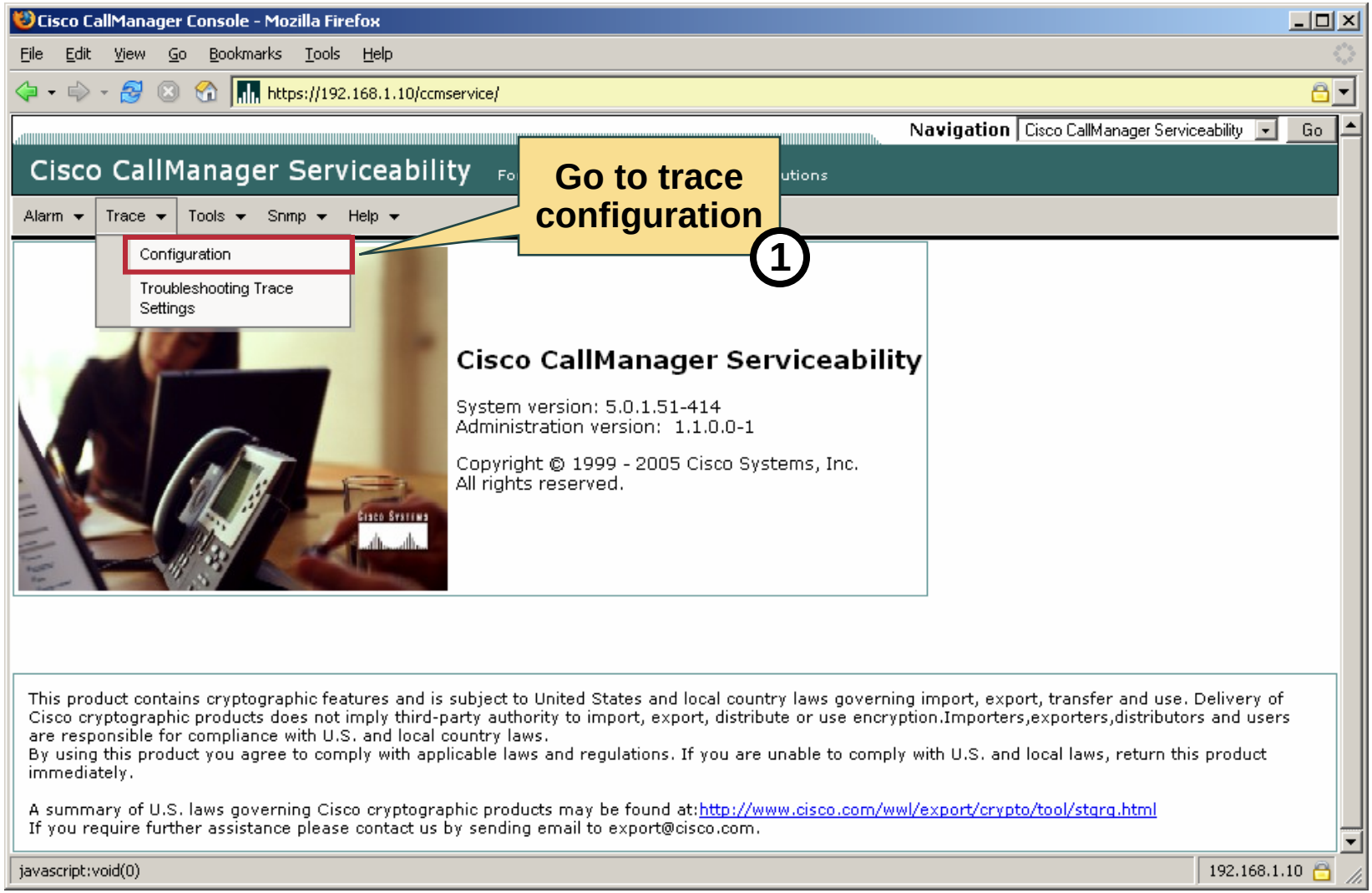
Cisco Unified CallManager 5.0 Traces



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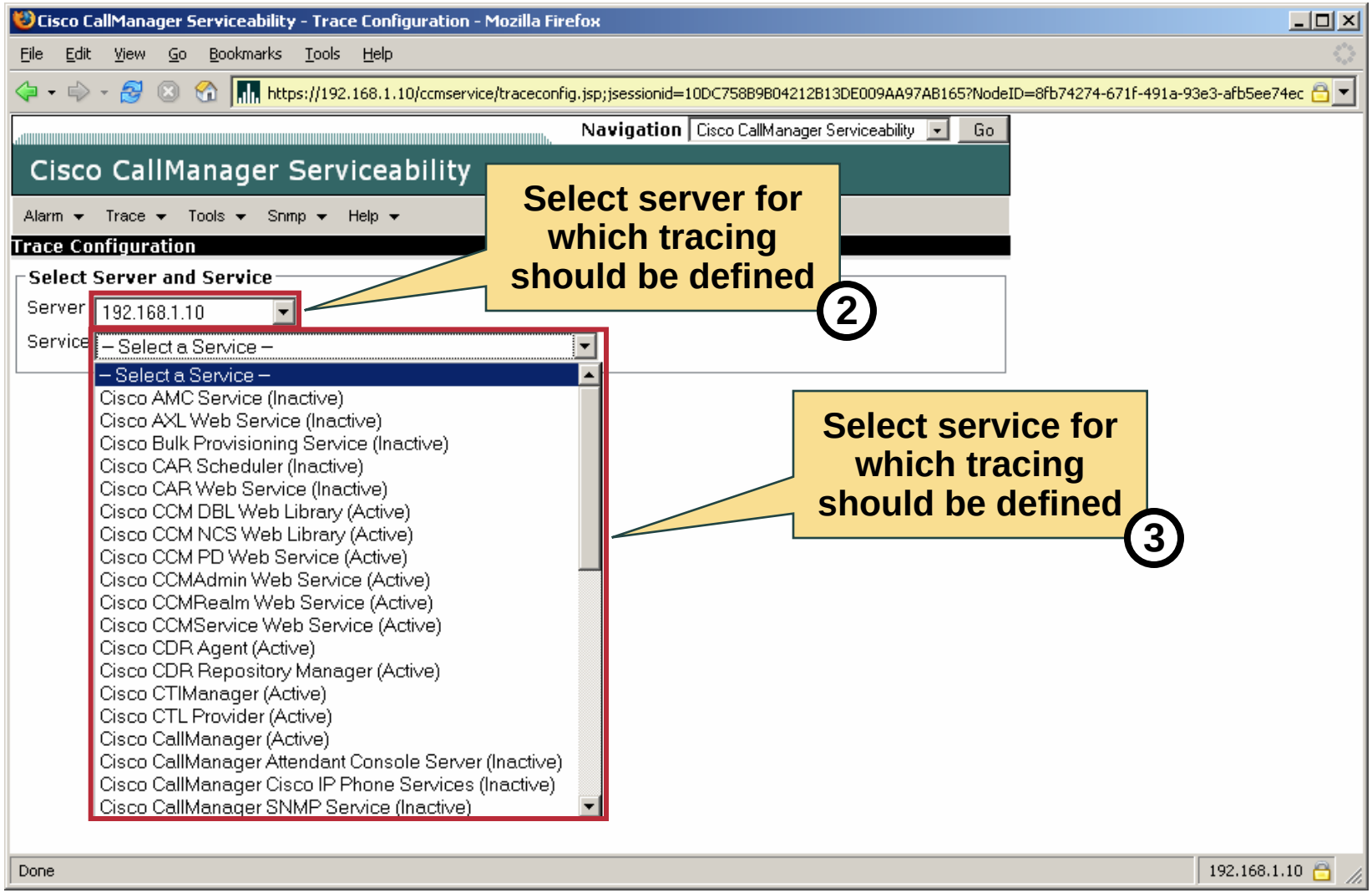
Cisco Unified CallManager 5.0 Traces



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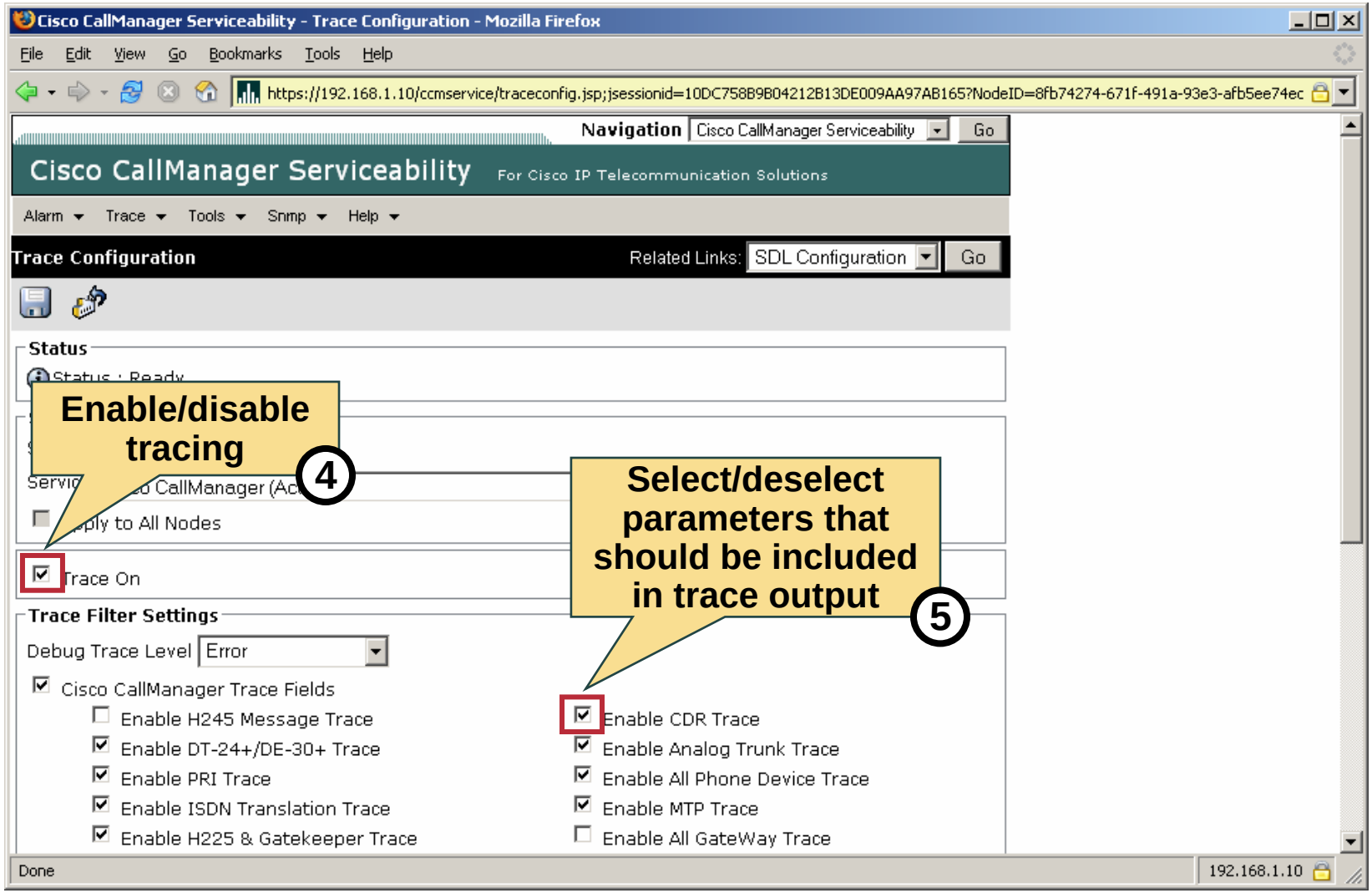
Cisco Unified CallManager 5.0 Traces



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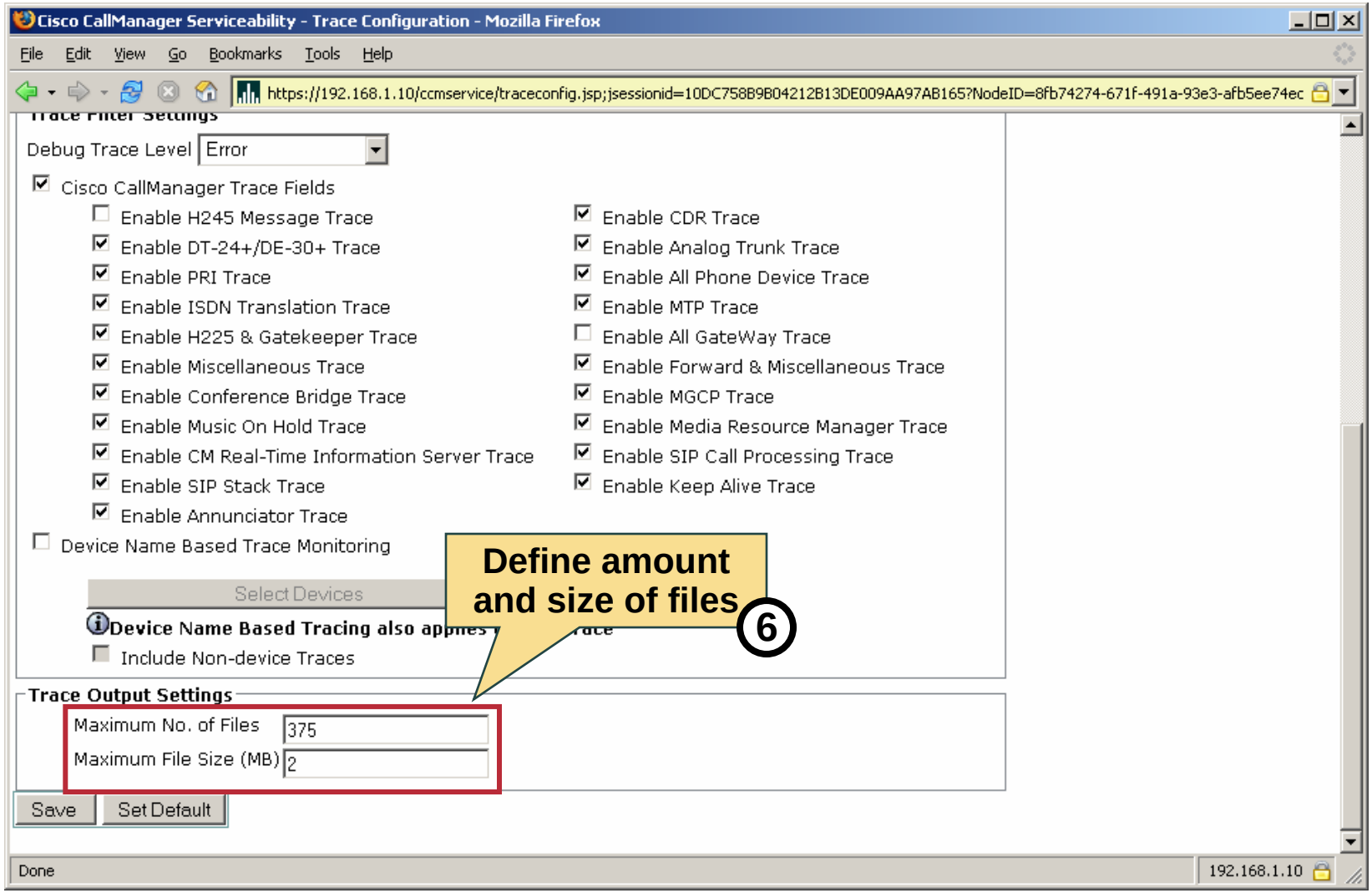
Cisco Unified CallManager 5.0 Traces



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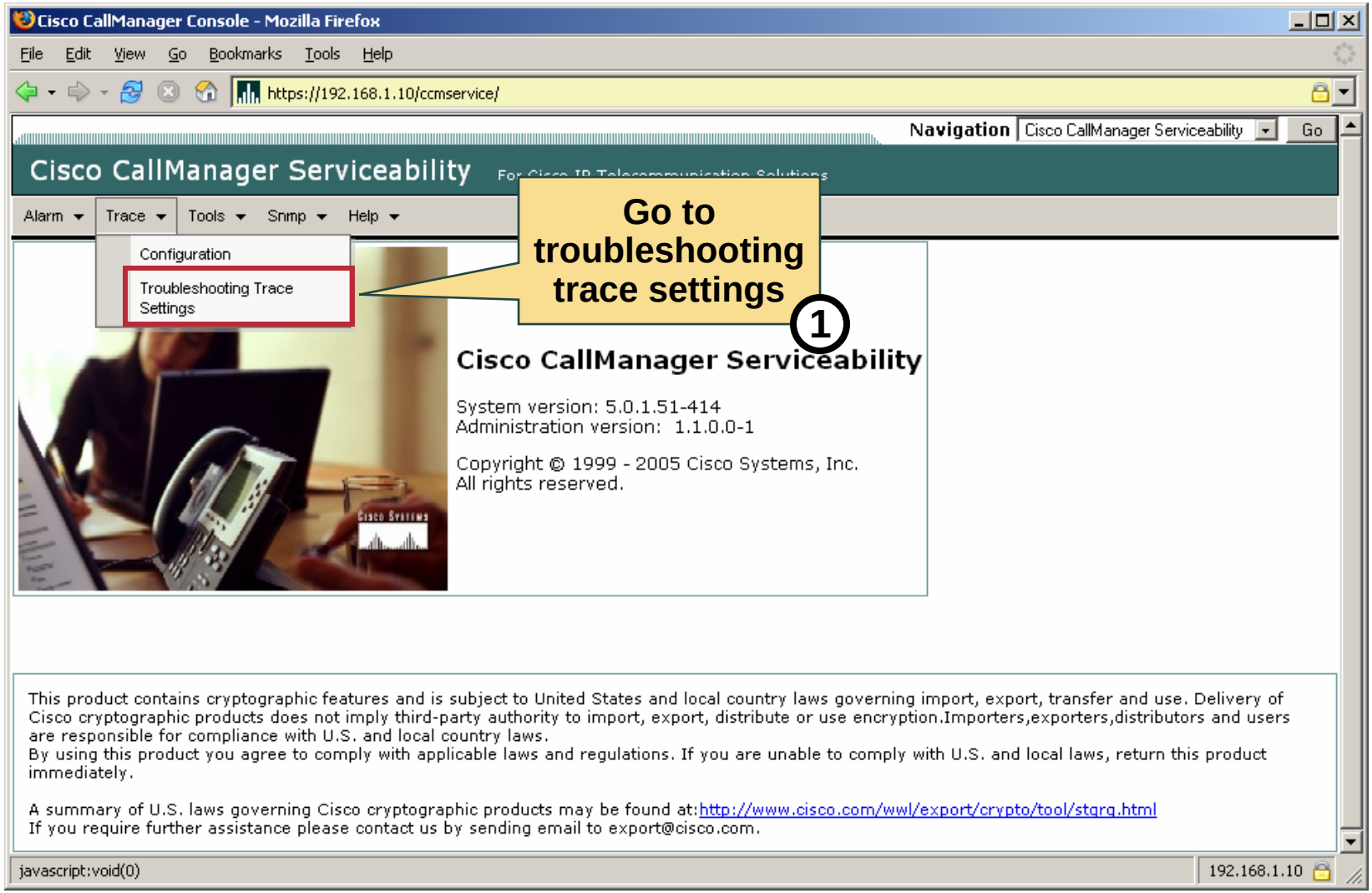
Cisco Unified CallManager 5.0 Traces



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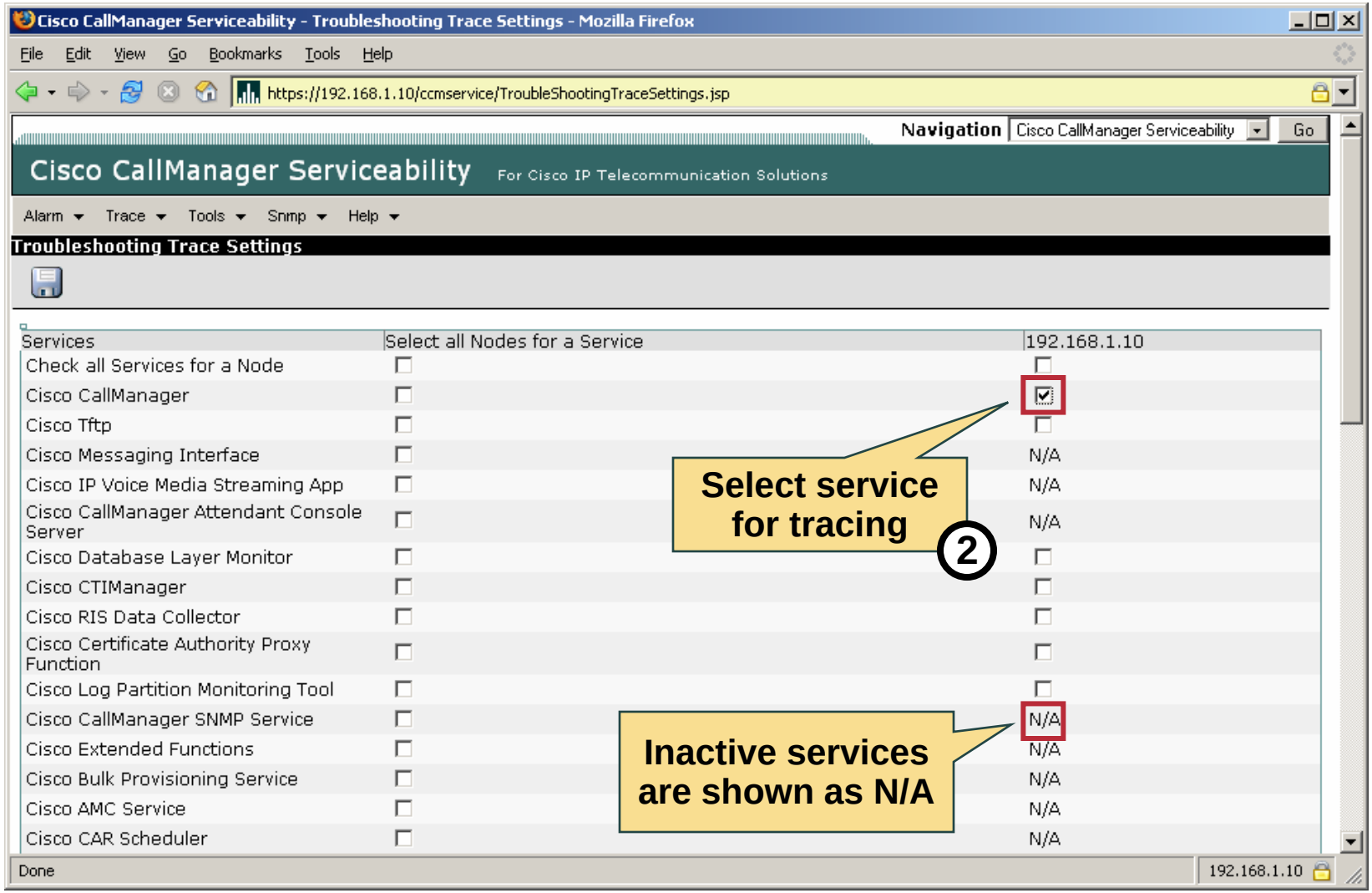
Cisco Unified CallManager 5.0 Traces



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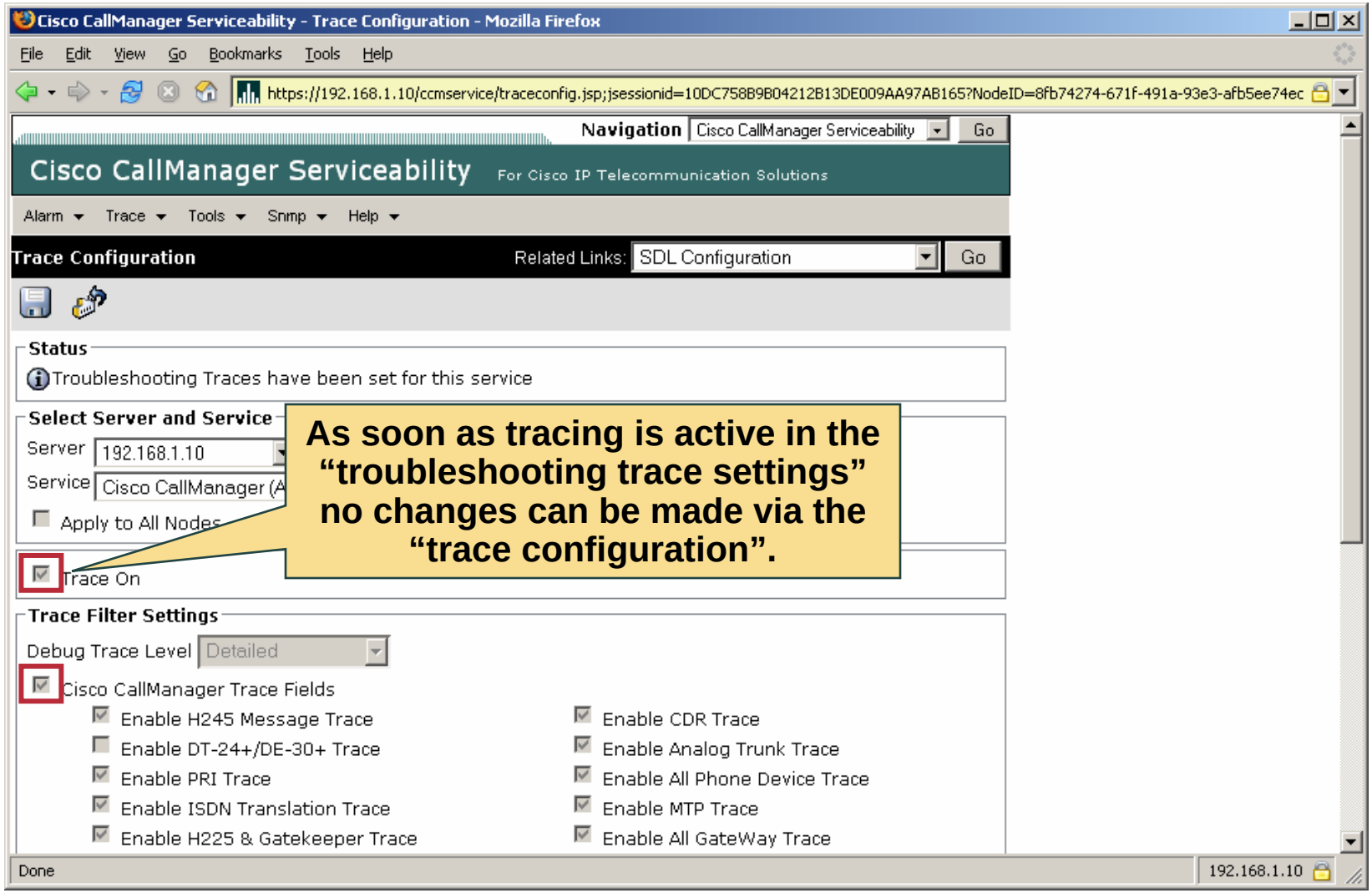
Cisco Unified CallManager 5.0 Traces



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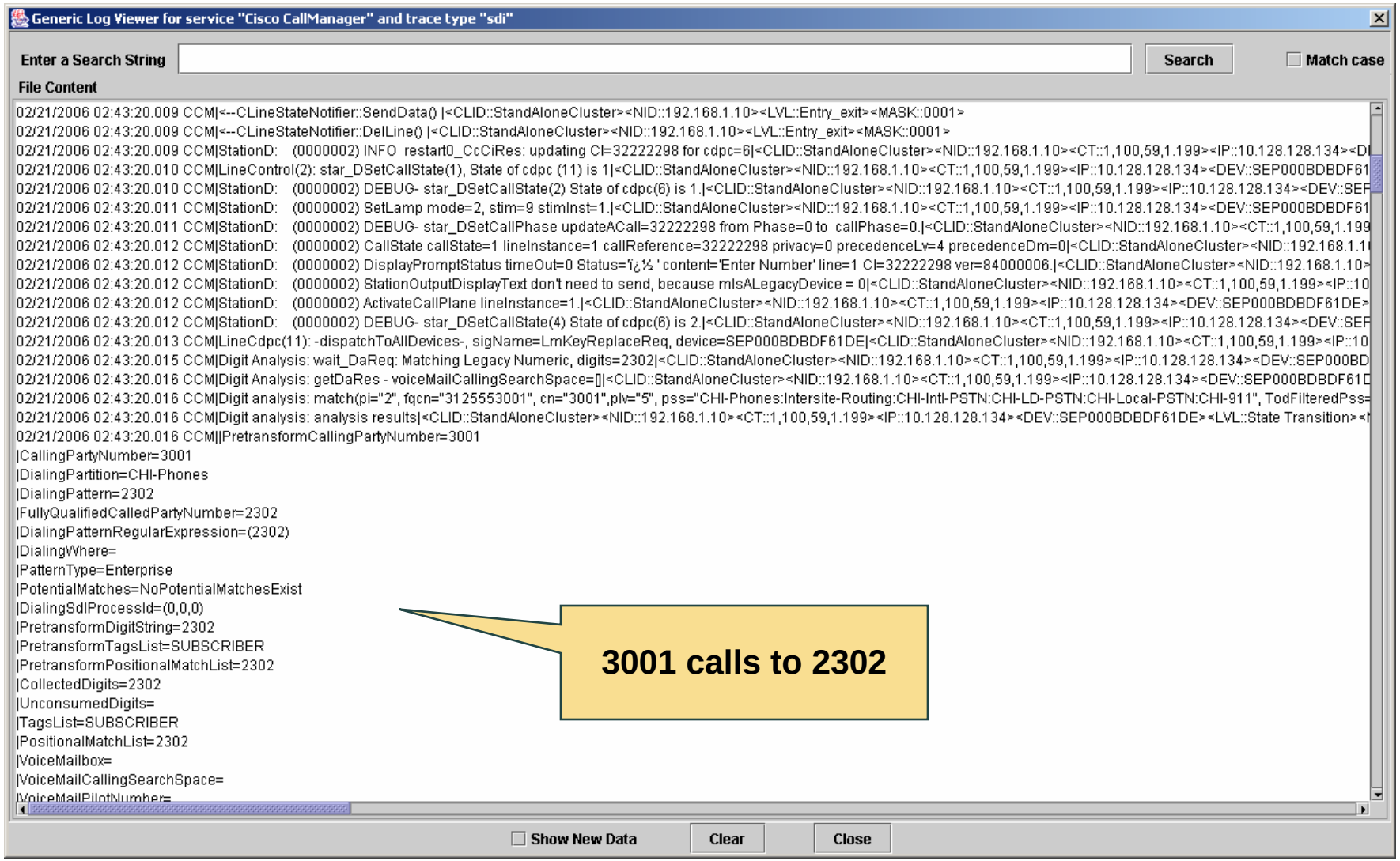
Cisco Unified CallManager 5.0 Traces



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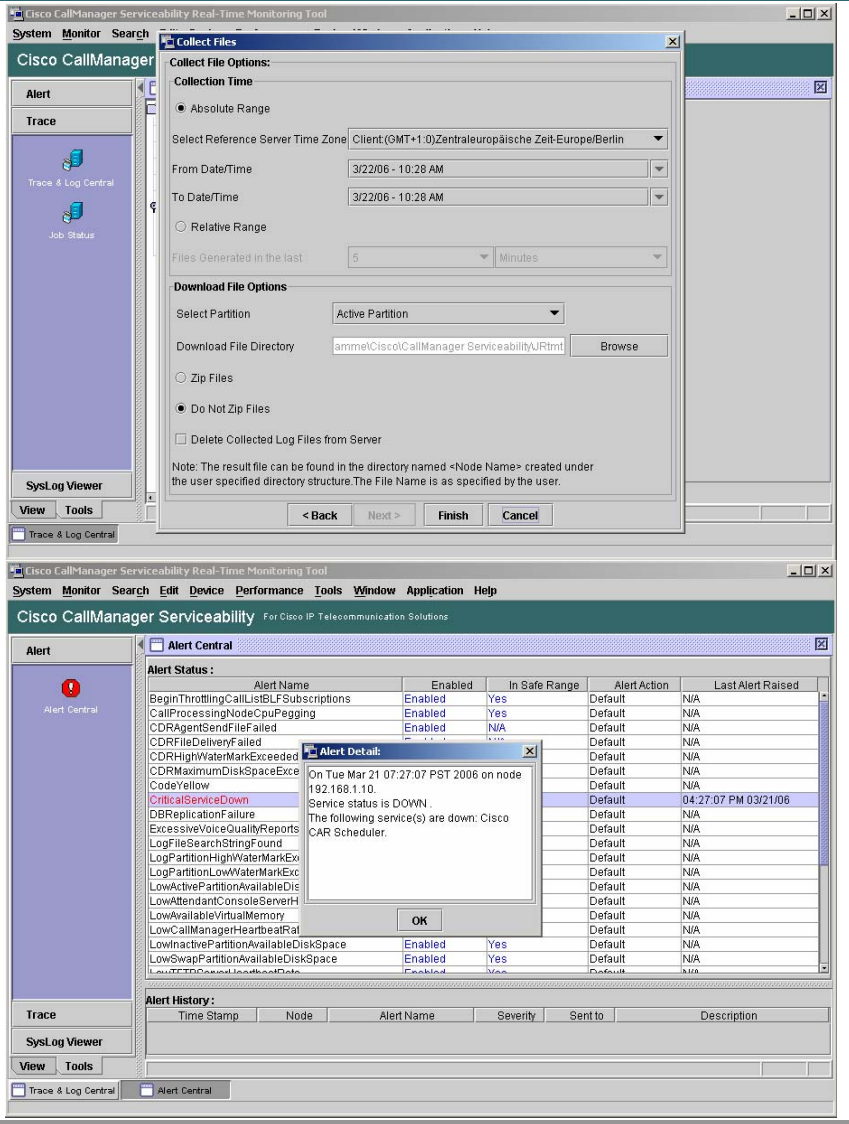
Cisco Unified CallManager 5.0 Traces: Trace Collection Tool

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



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Tools Menu

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

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

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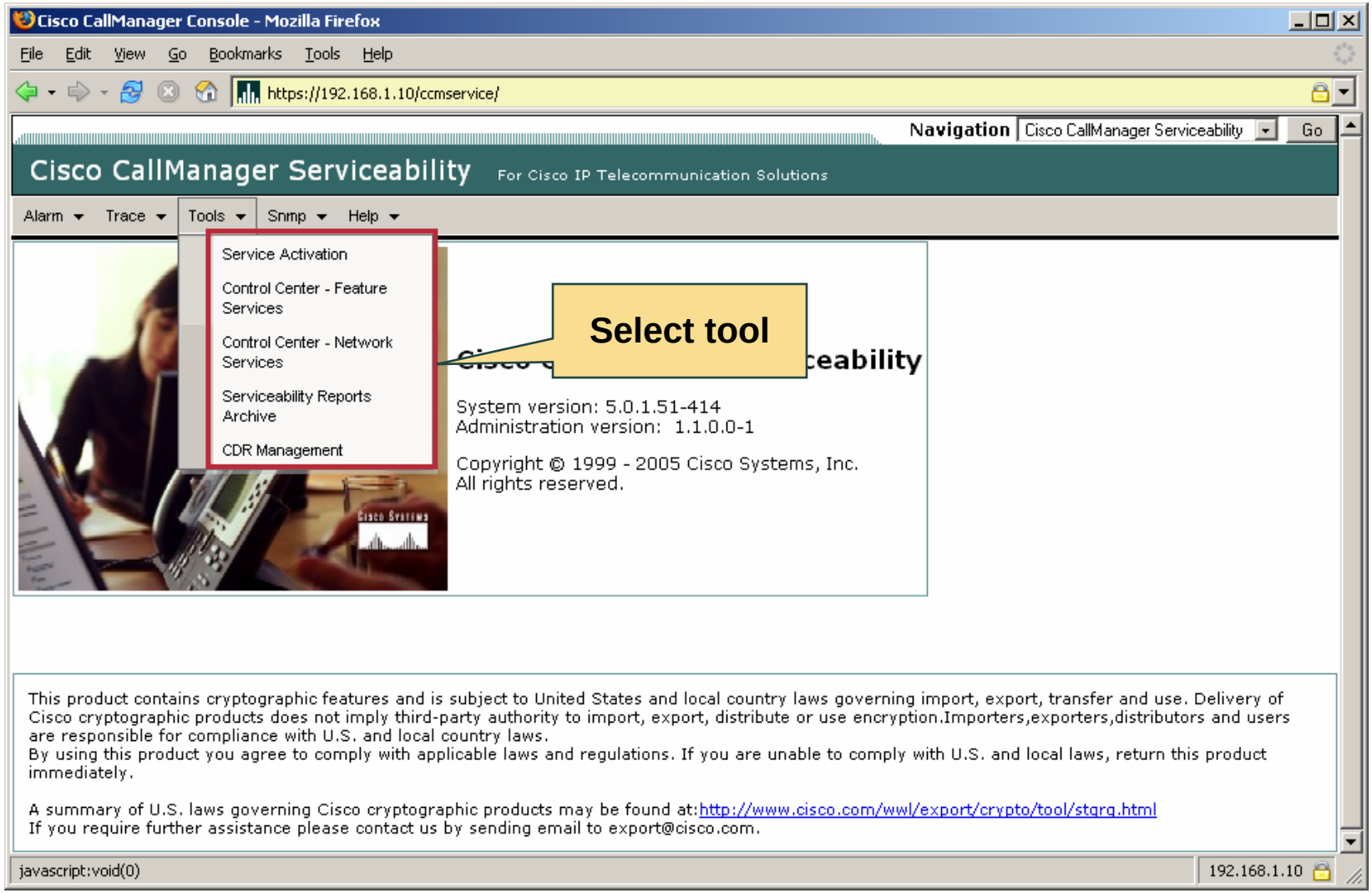
Tools Menu



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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.**

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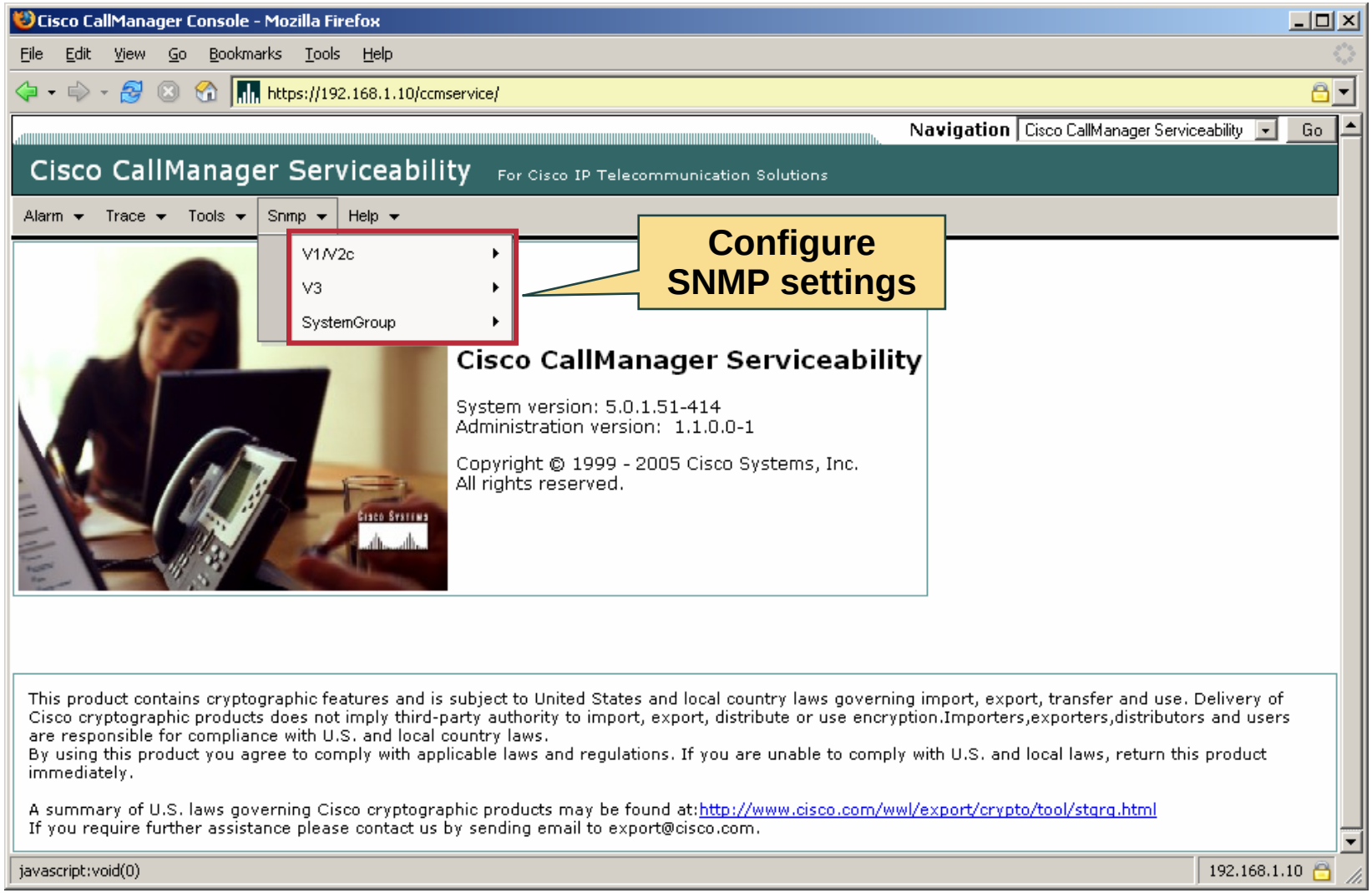
SNMP Configuration



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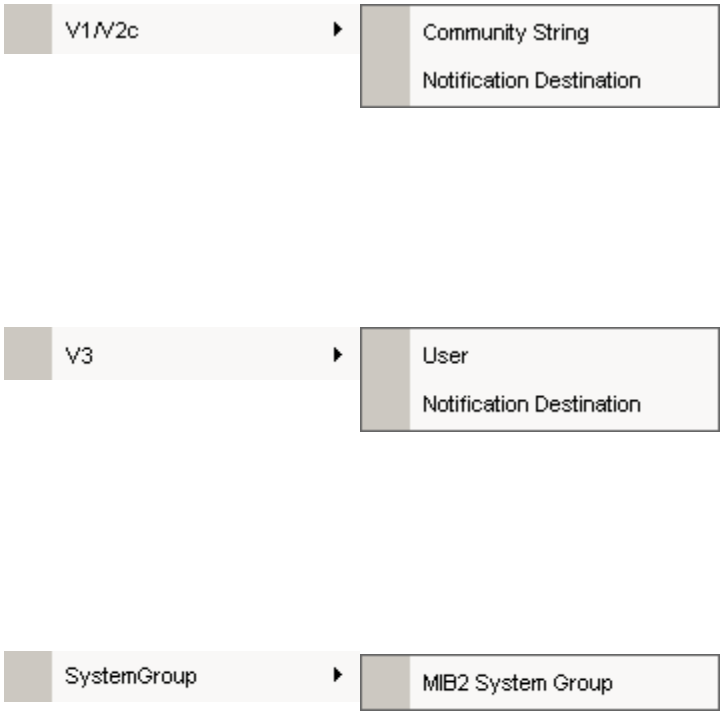
SNMP Configuration

- **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.**



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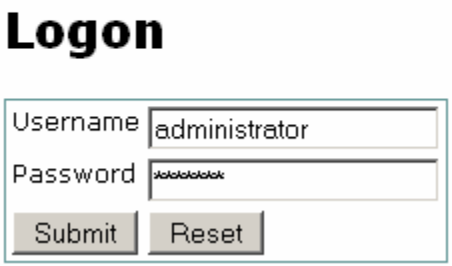
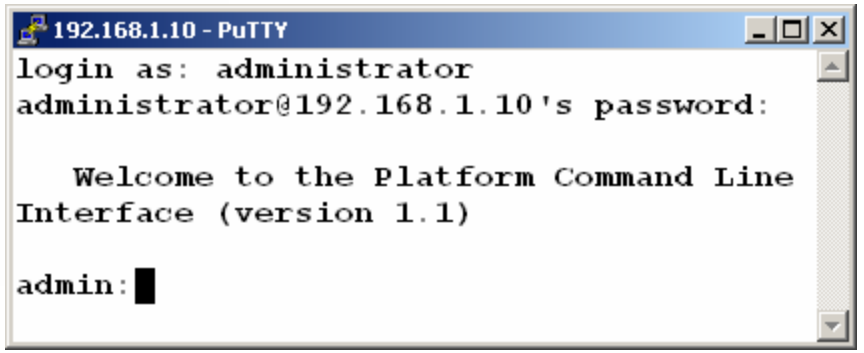
Platform Administration

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

- **Platform administration web GUI**
- **Platform administration CLI**

Username and password are case sensitive





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Platform Administration

CLI commands:

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

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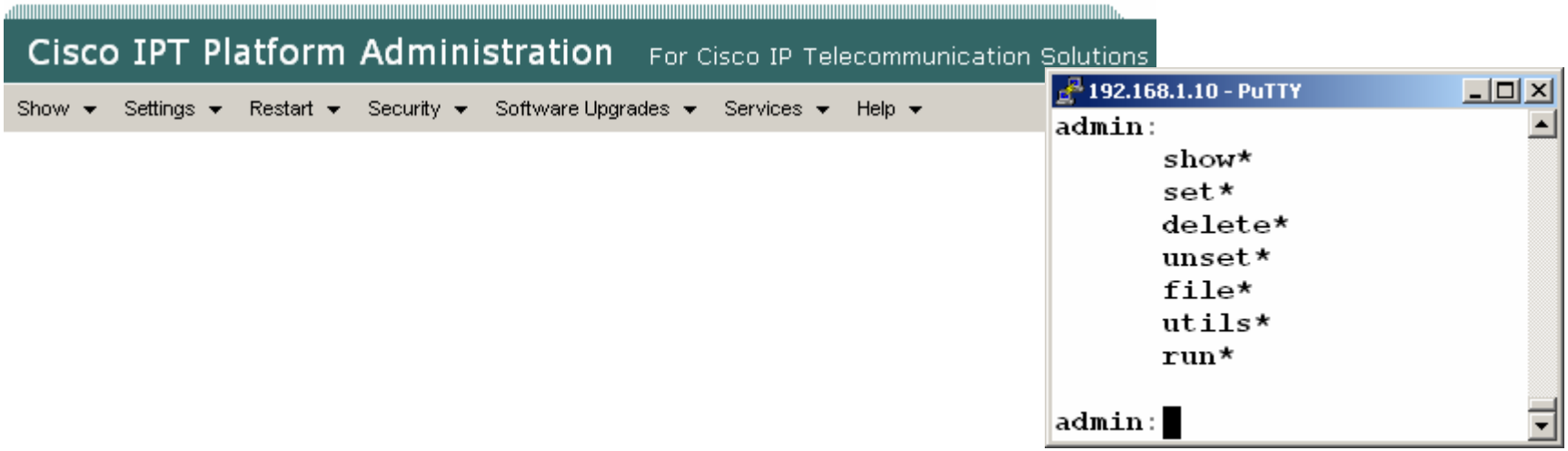
Platform Administration

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.**



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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.

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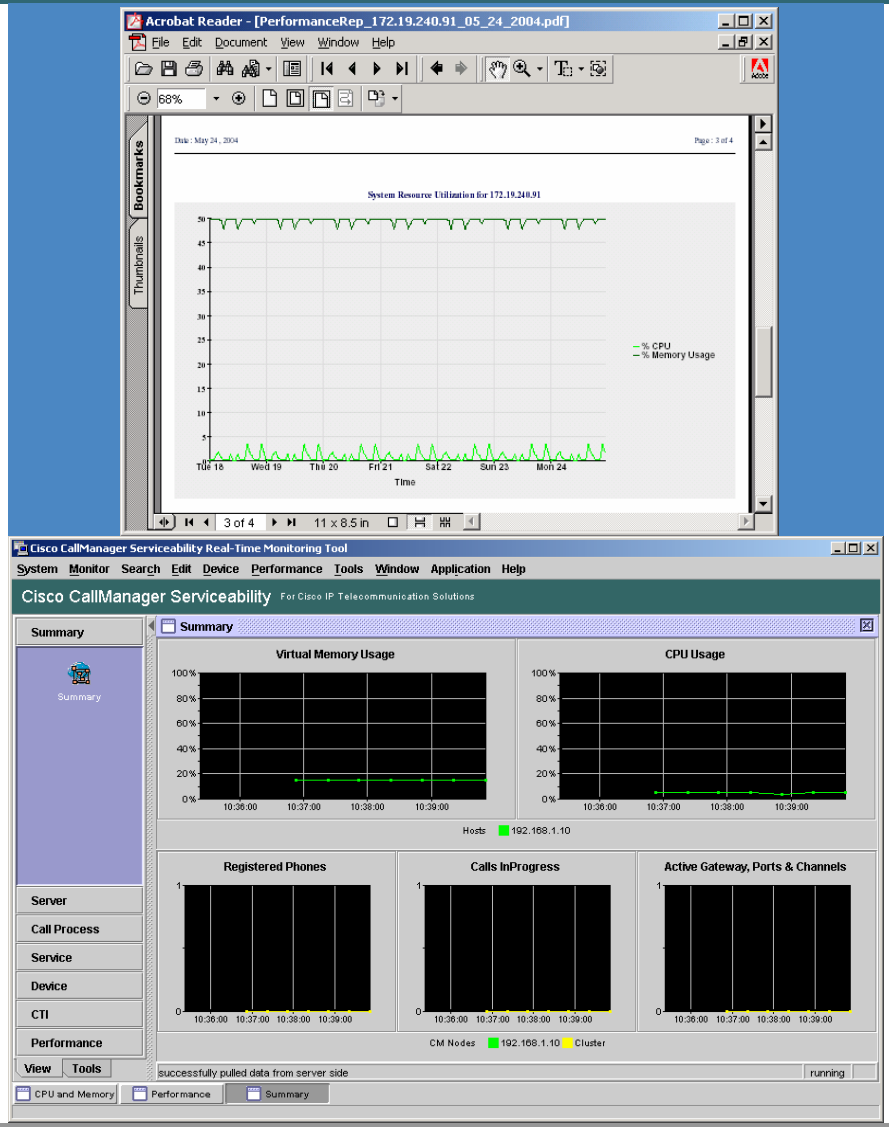
Additional Tools to Manage Cisco Unified CallManager 5.0: RTMT

- 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)



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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.

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Administering Cisco Unified CallManager 5.0

Backing Up and Restoring Cisco Unified CallManager 5.0

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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)**

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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 backups 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).

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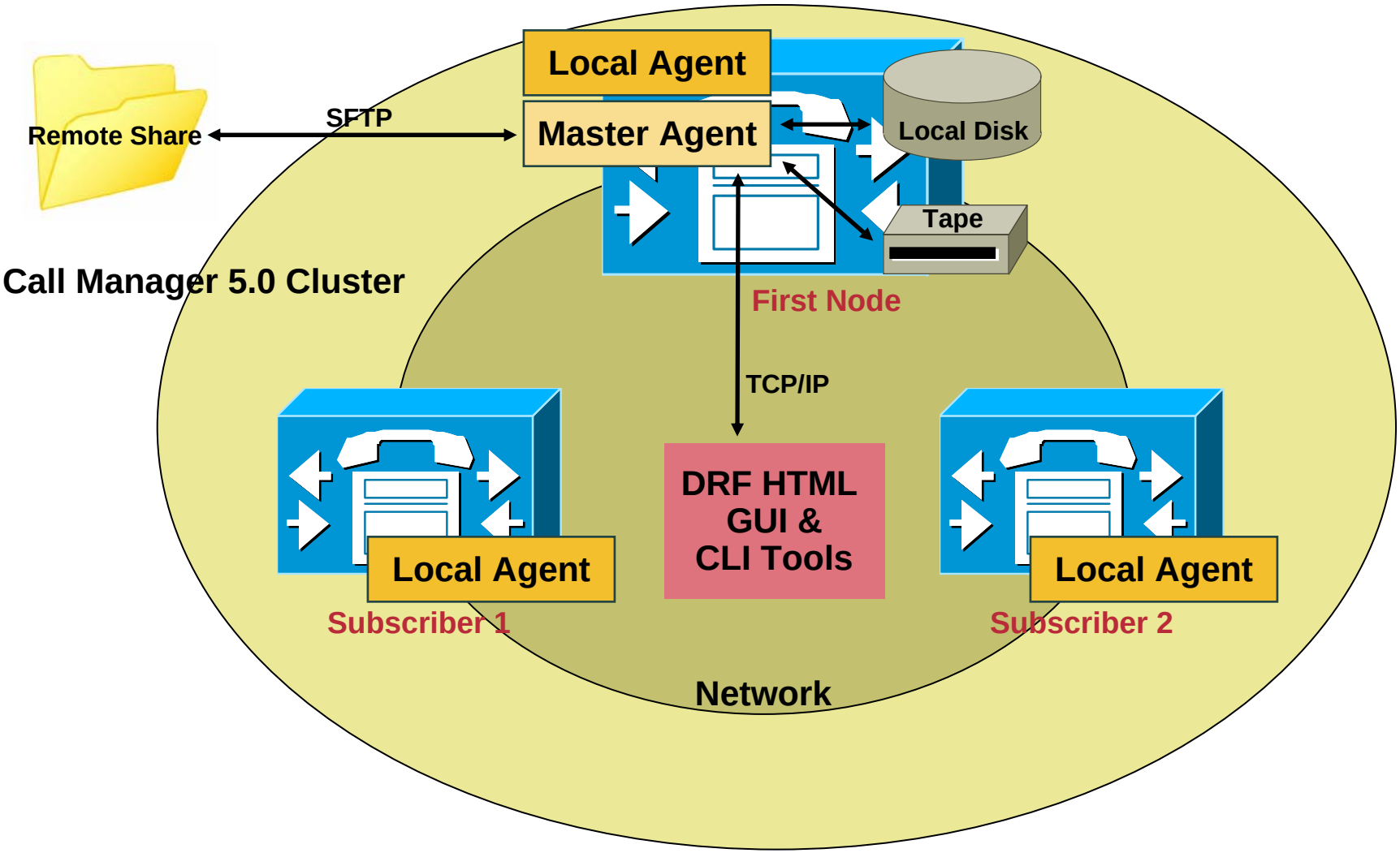
CallManager 5.0 Architecture



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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**

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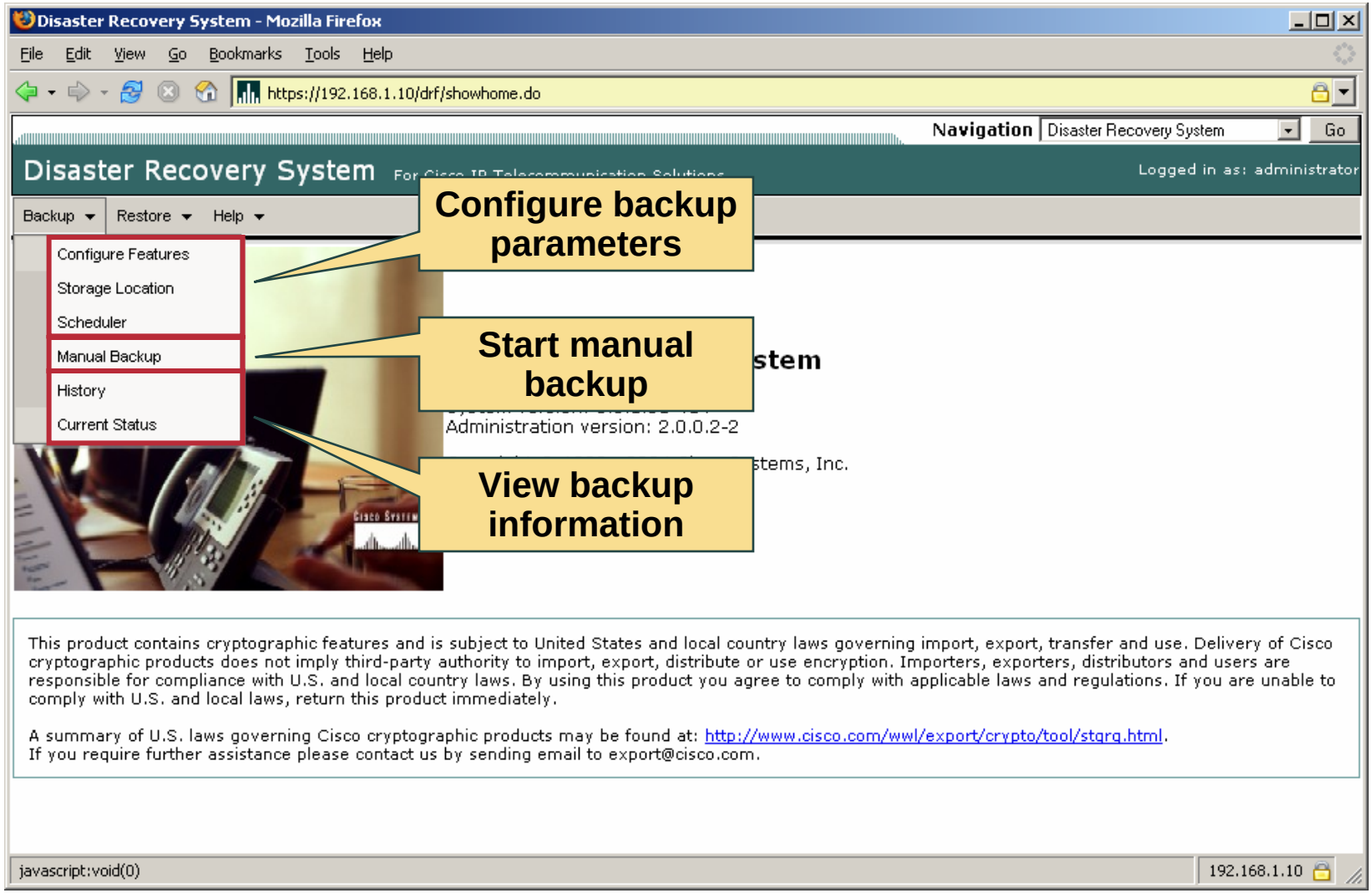
Backup Menu



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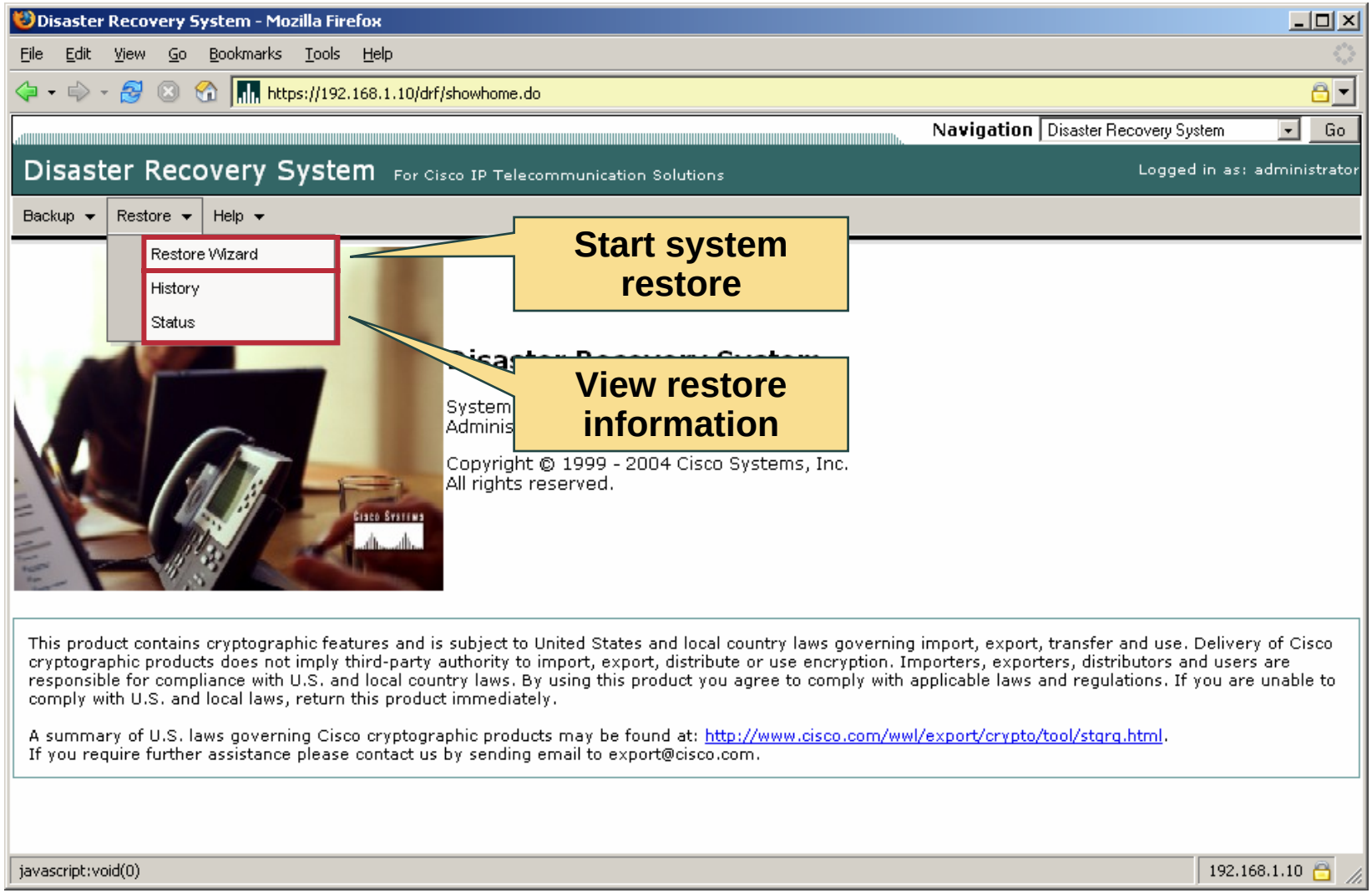
Restore Menu



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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.**

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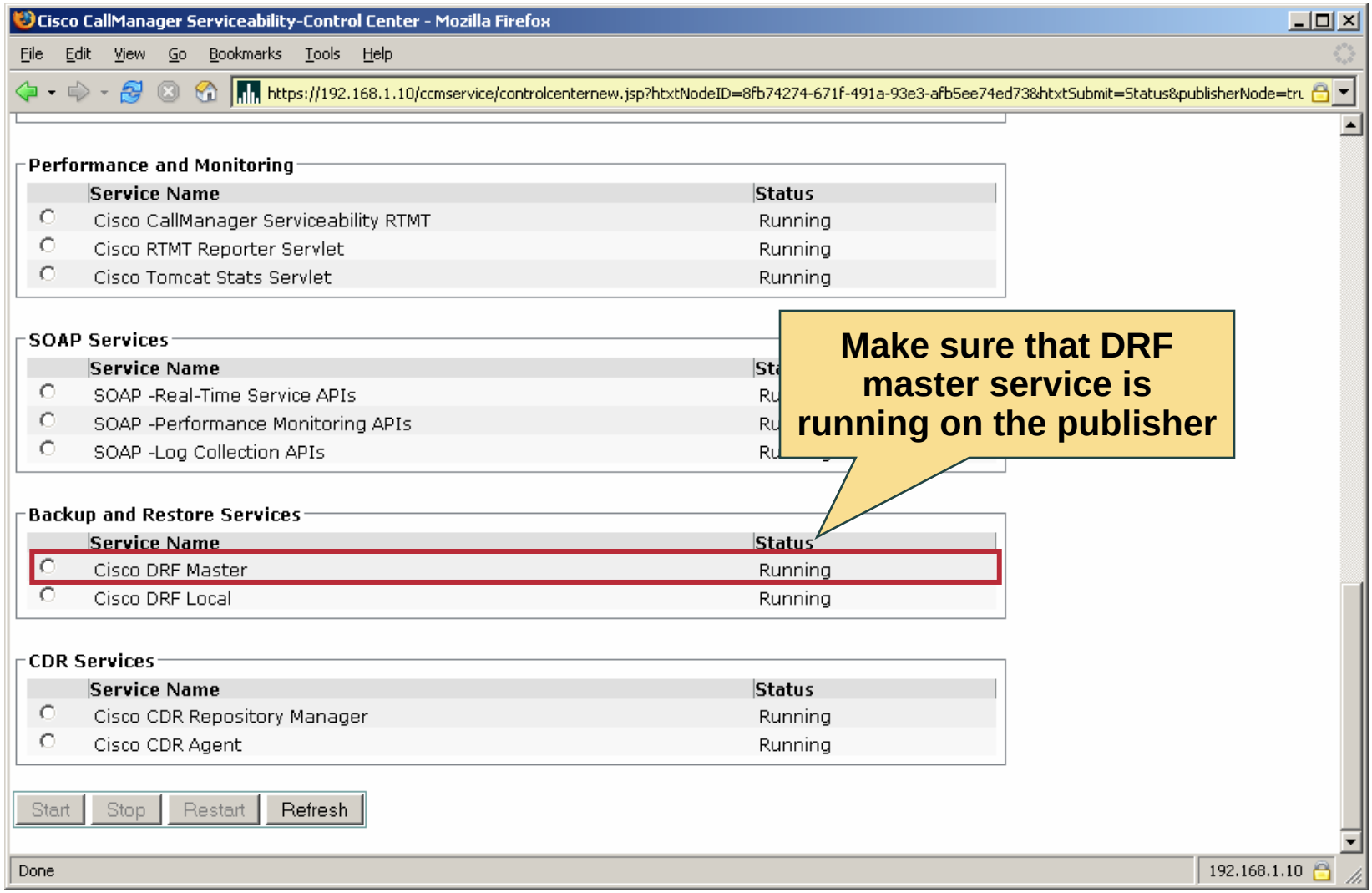
Backup and Restore Procedure: Service Verification



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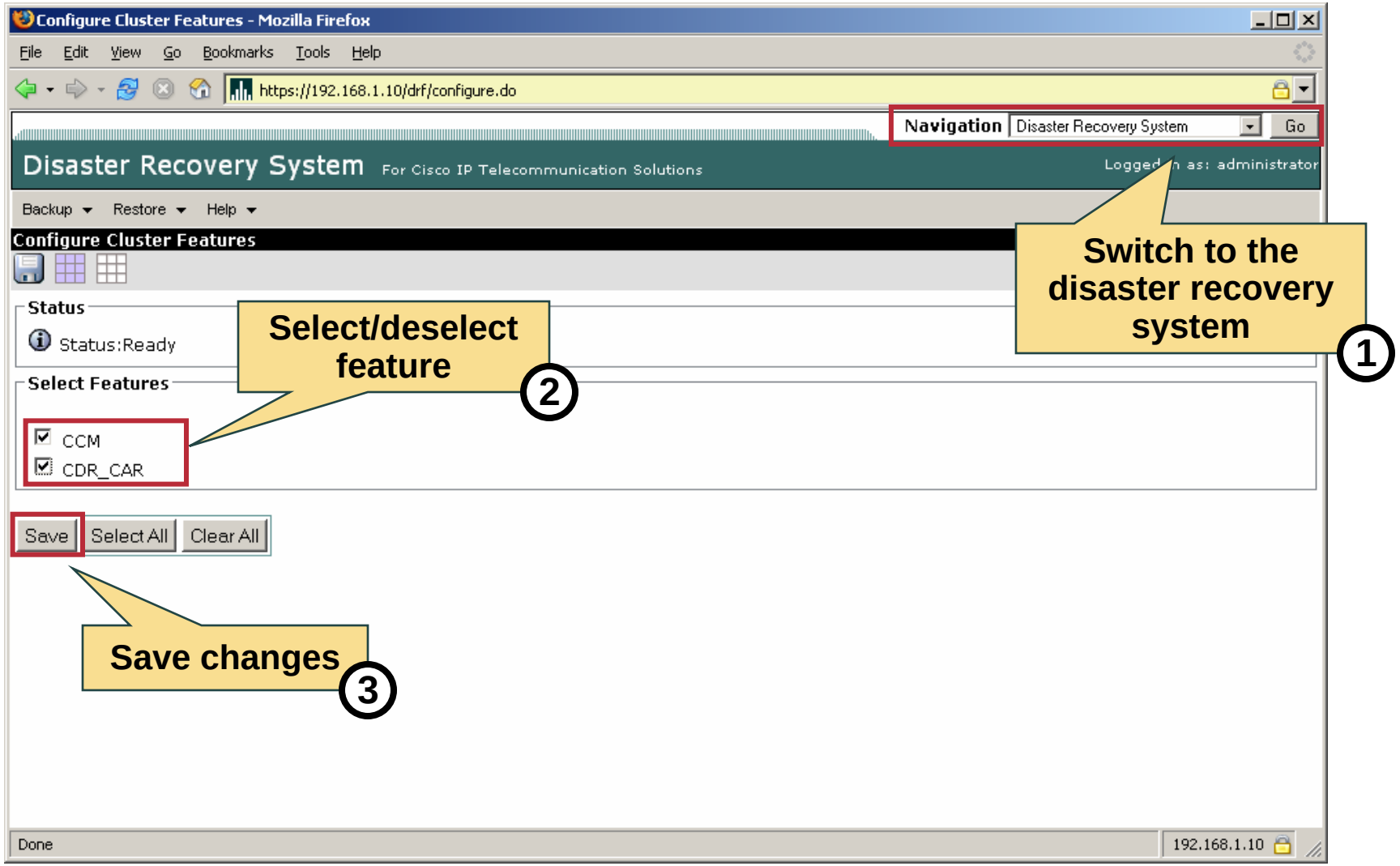
Backup and Restore Procedure: Feature Backup Selection



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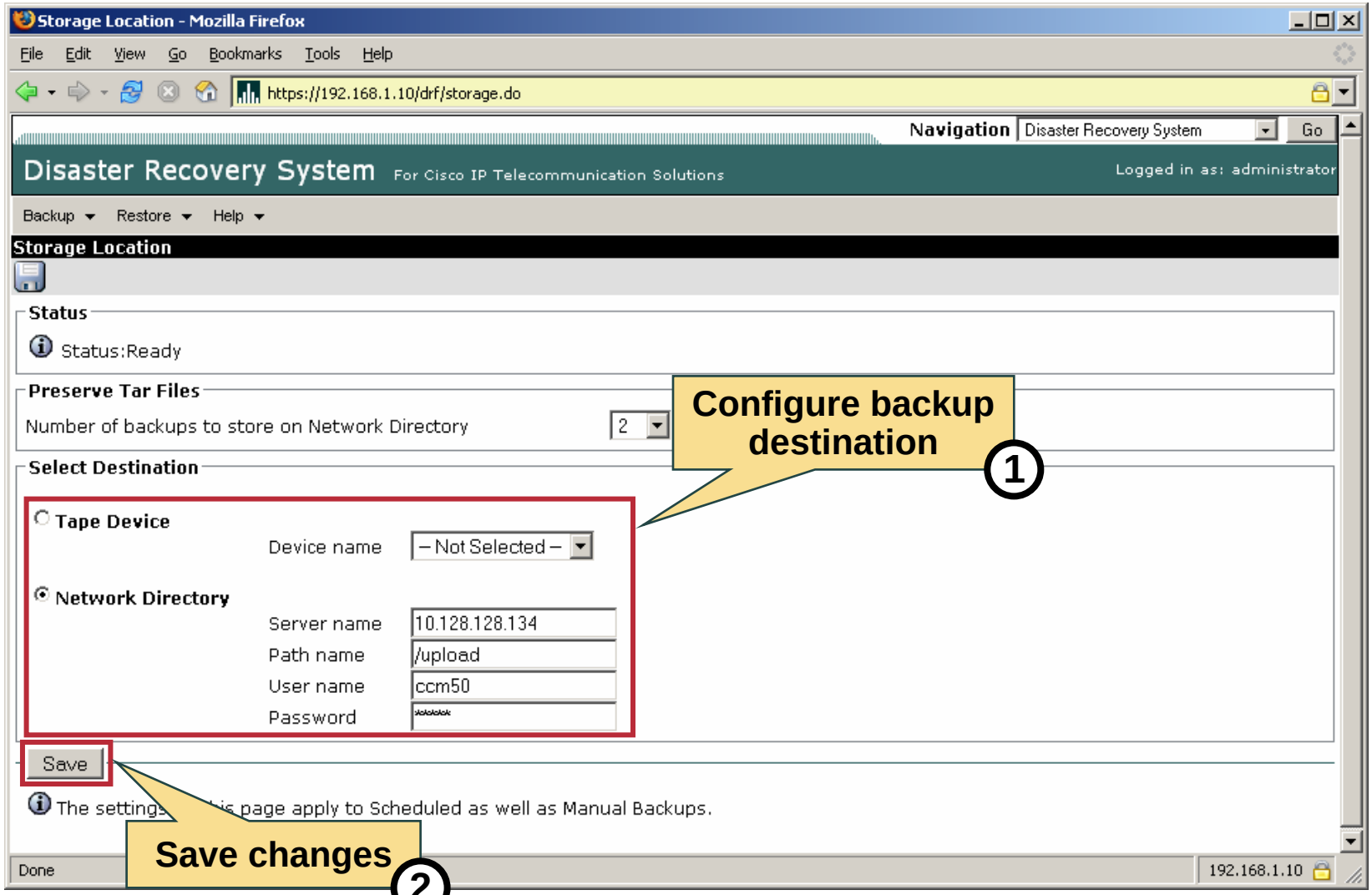
Backup and Restore Procedure: Backup Target Configuration



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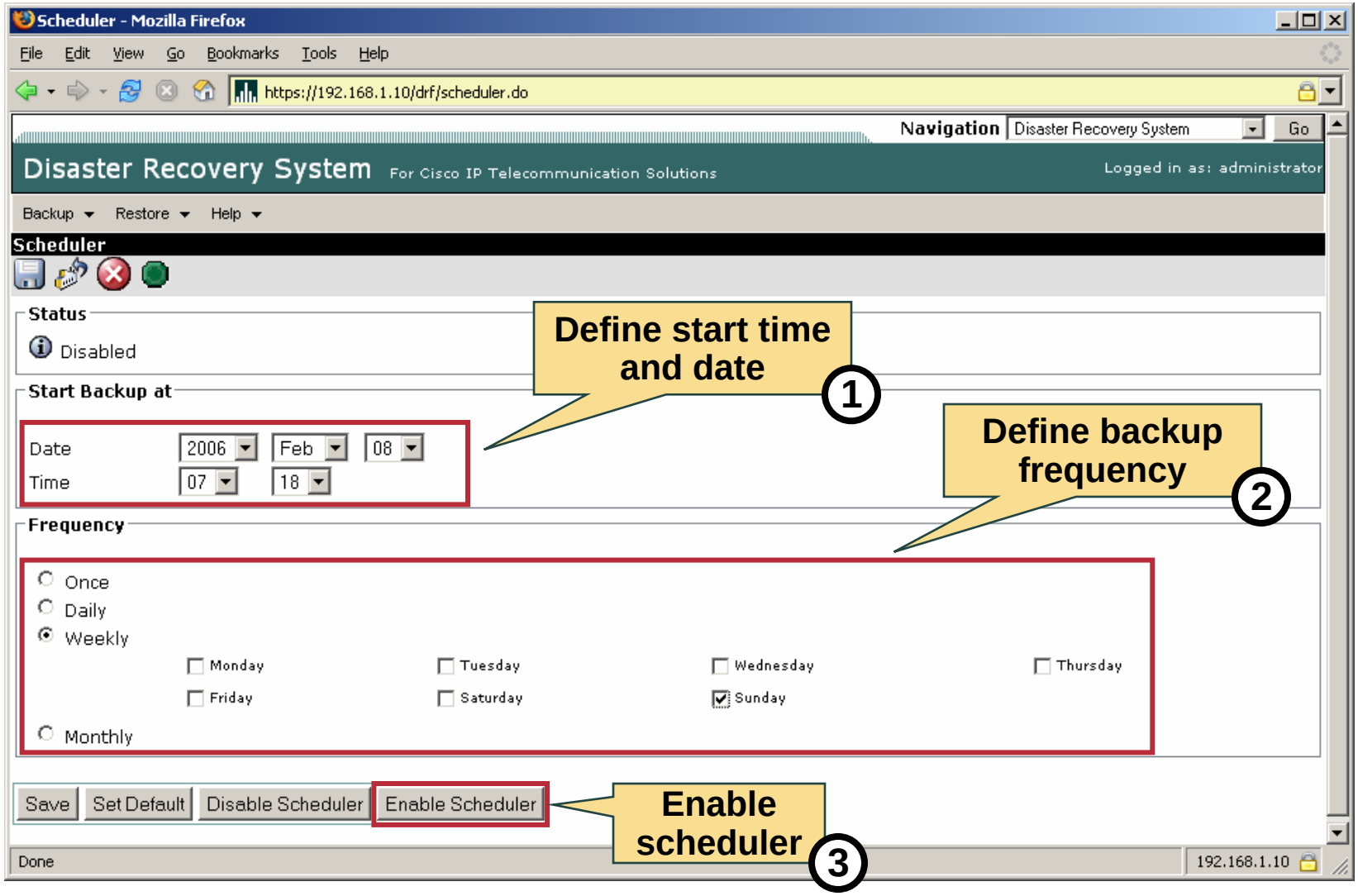
Backup and Restore Procedure: Backup Scheduler Configuration



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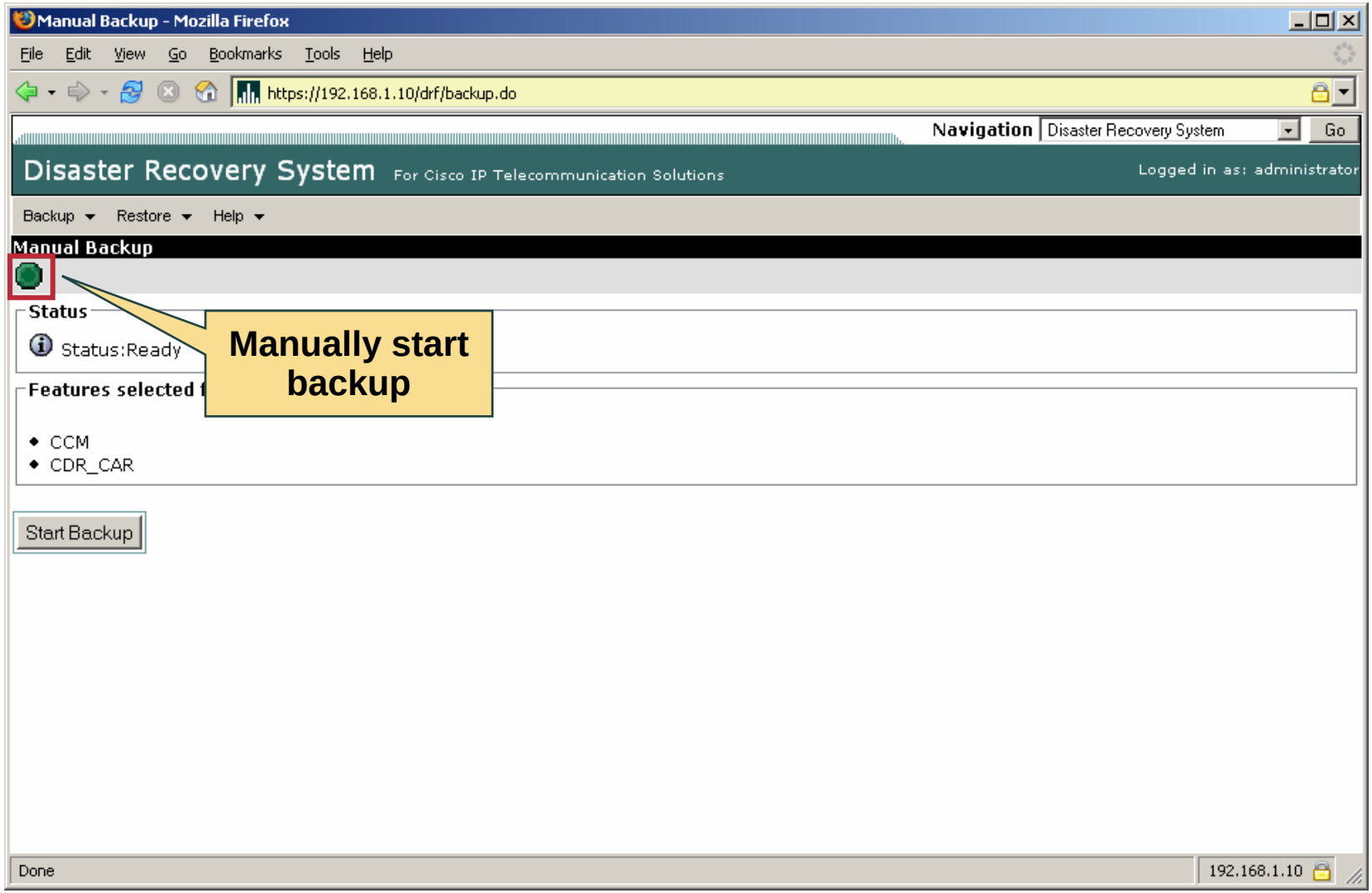
Backup and Restore Procedure: Manual Backup



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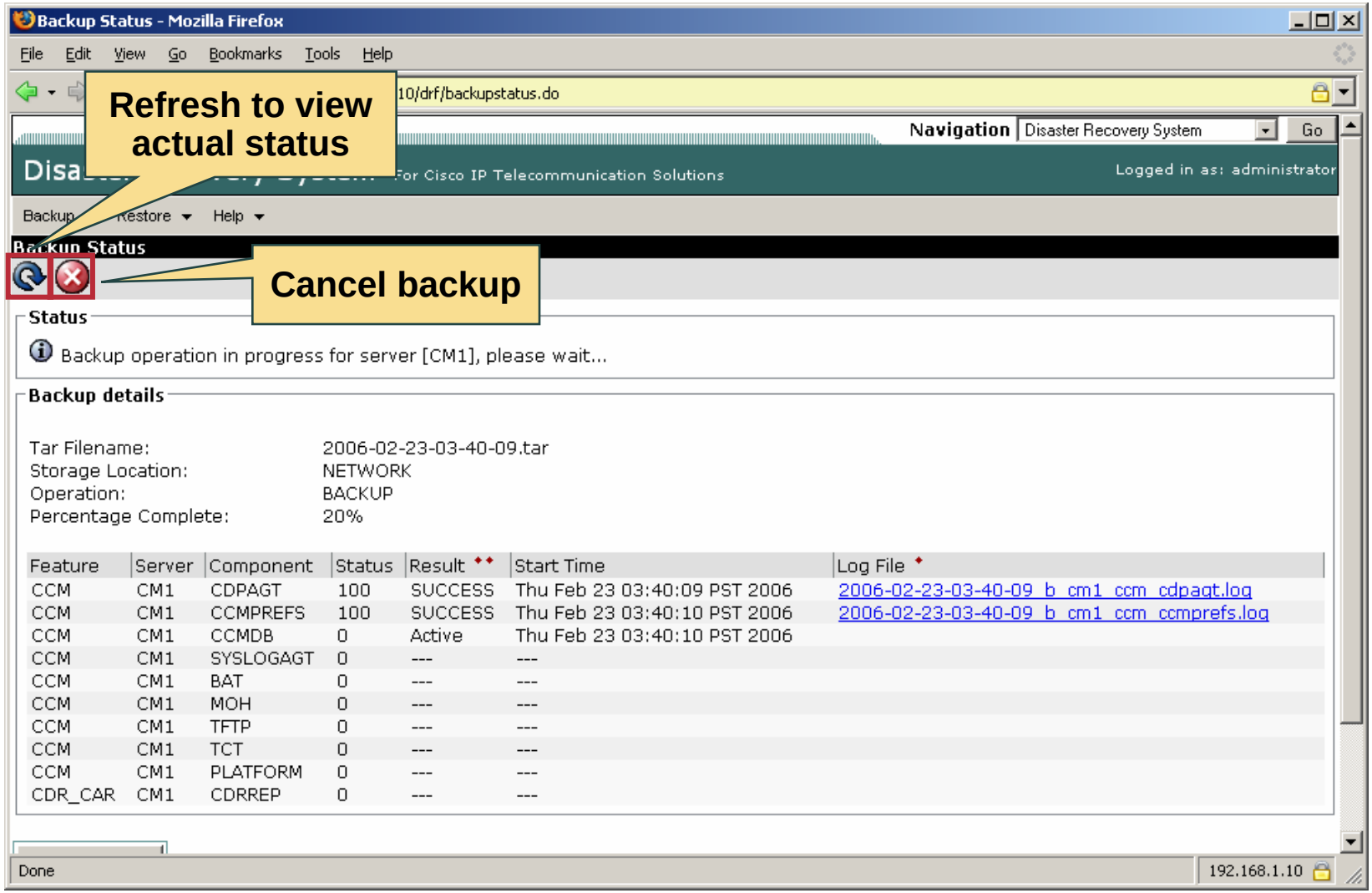
Backup and Restore Procedure: Manual Backup



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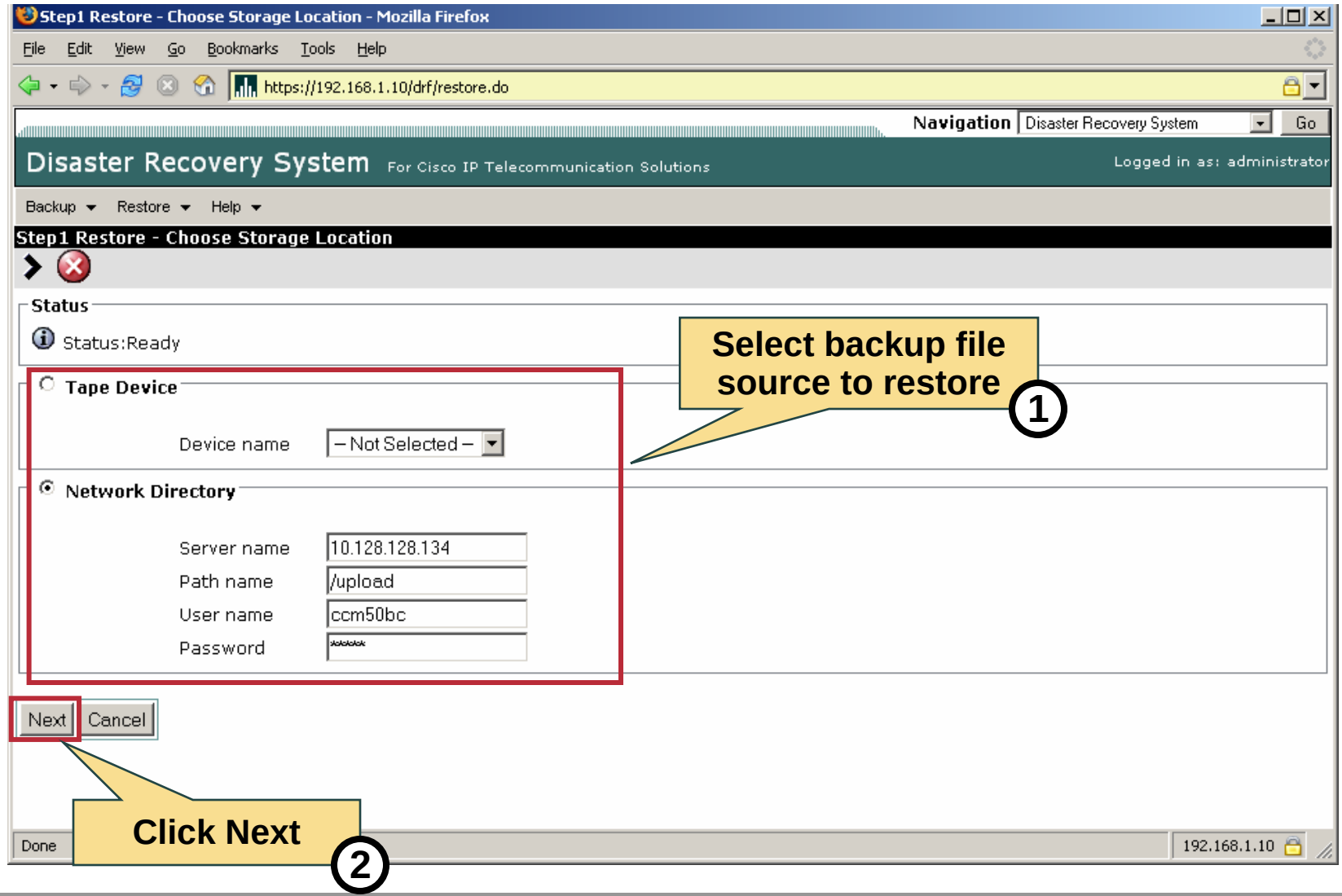
Backup and Restore Procedure: Restore Wizard



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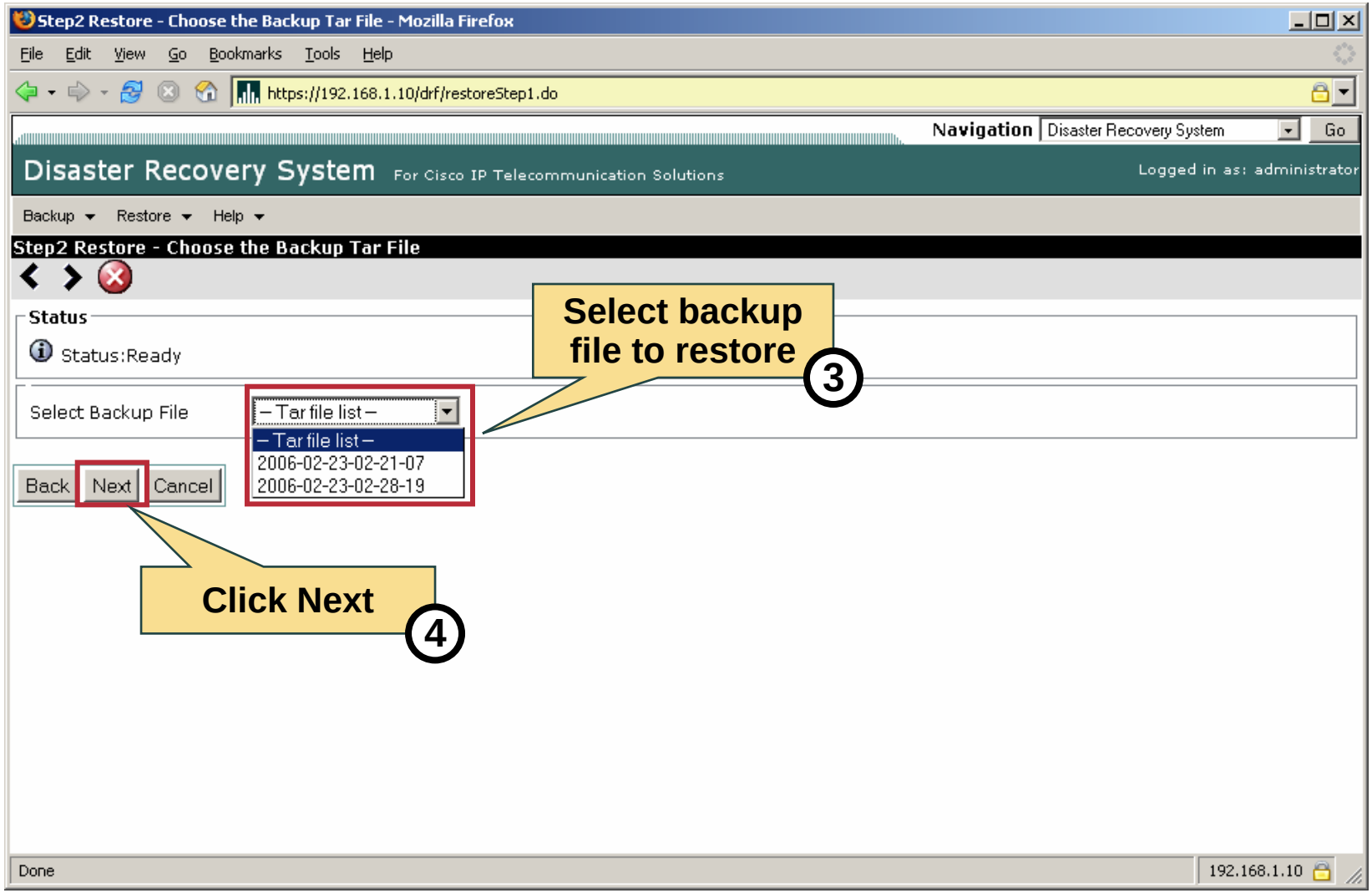
Backup and Restore Procedure: Restore Wizard



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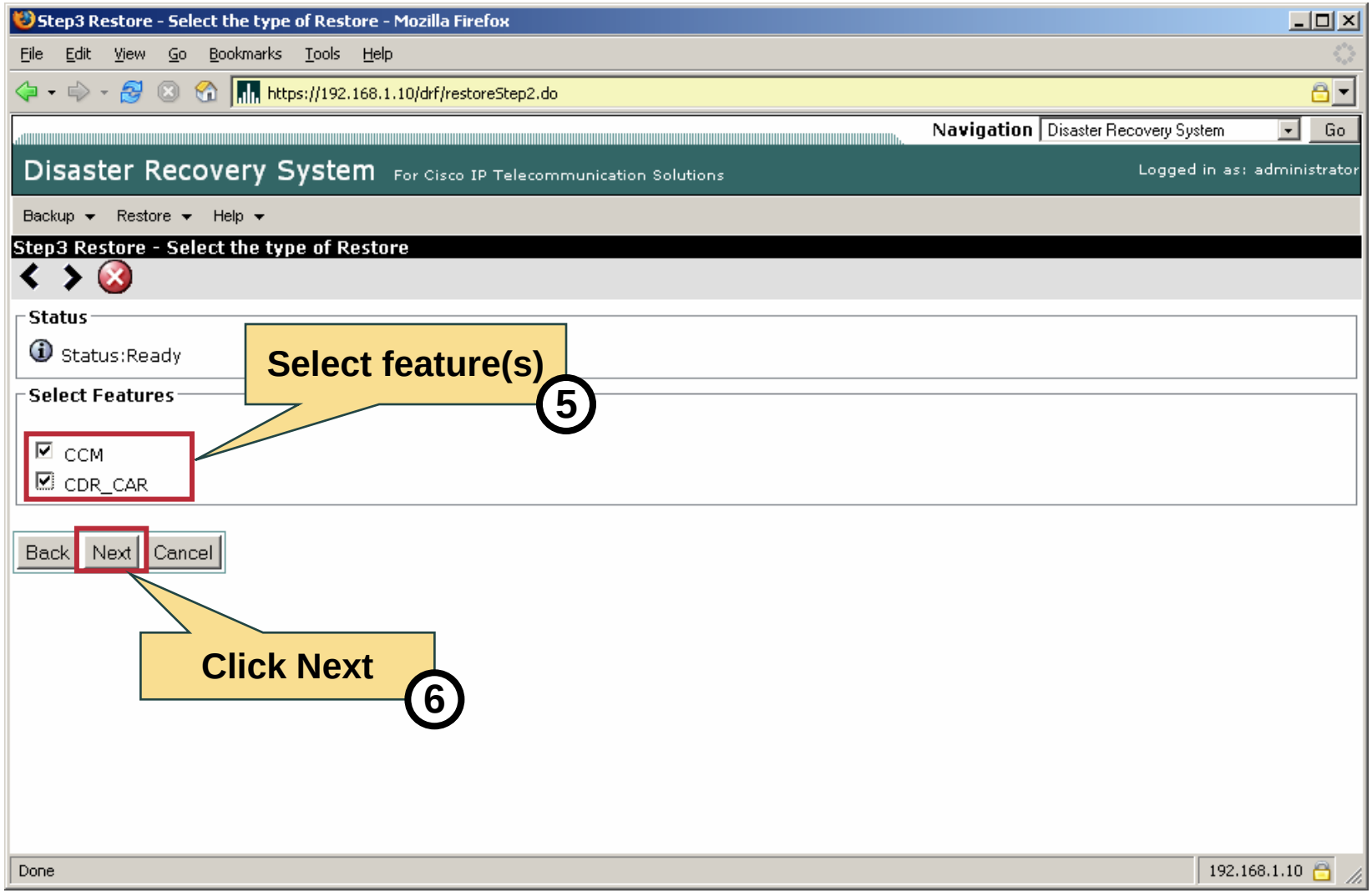
Backup and Restore Procedure: Restore Wizard



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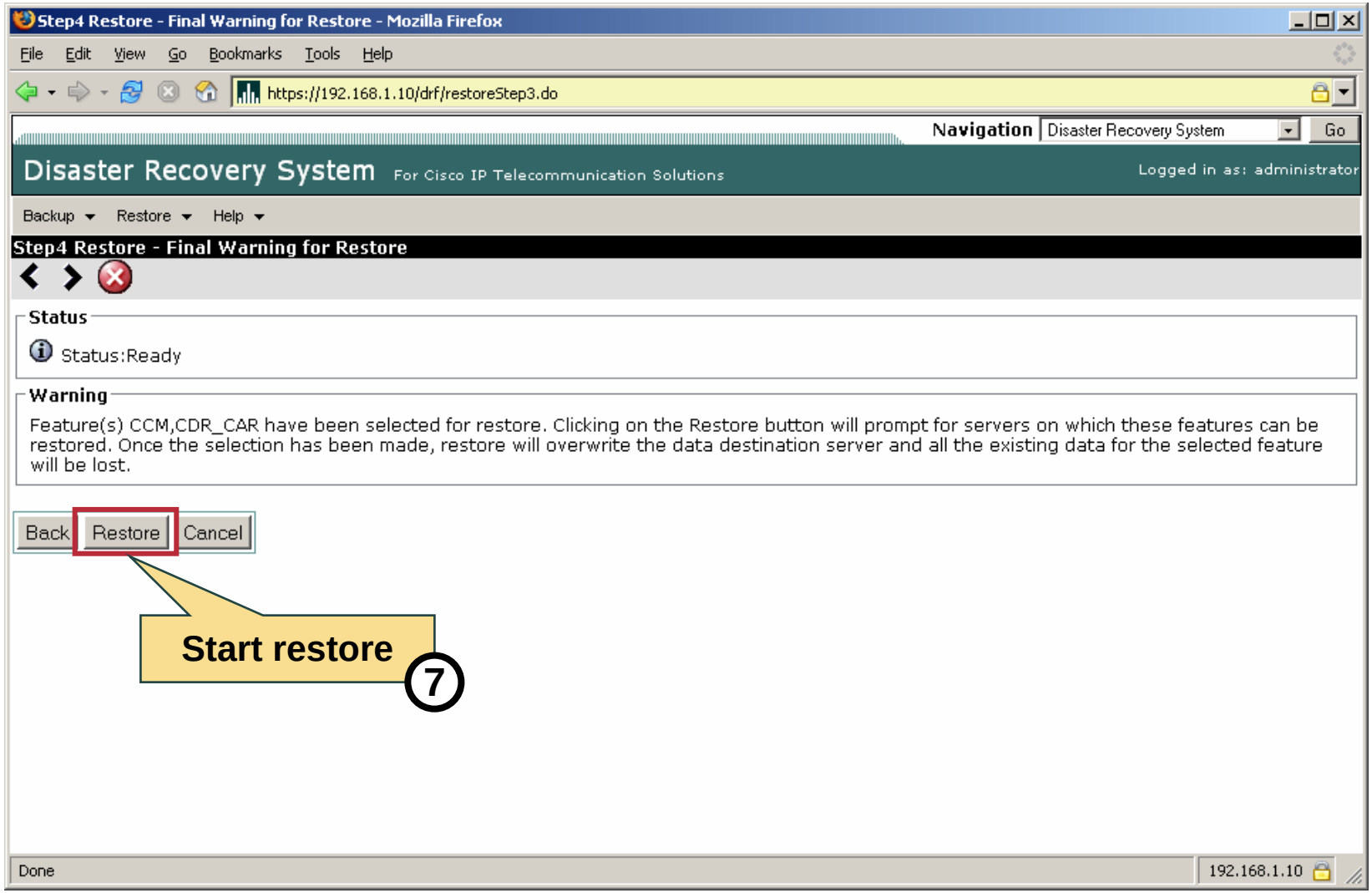
Backup and Restore Procedure: Restore Wizard



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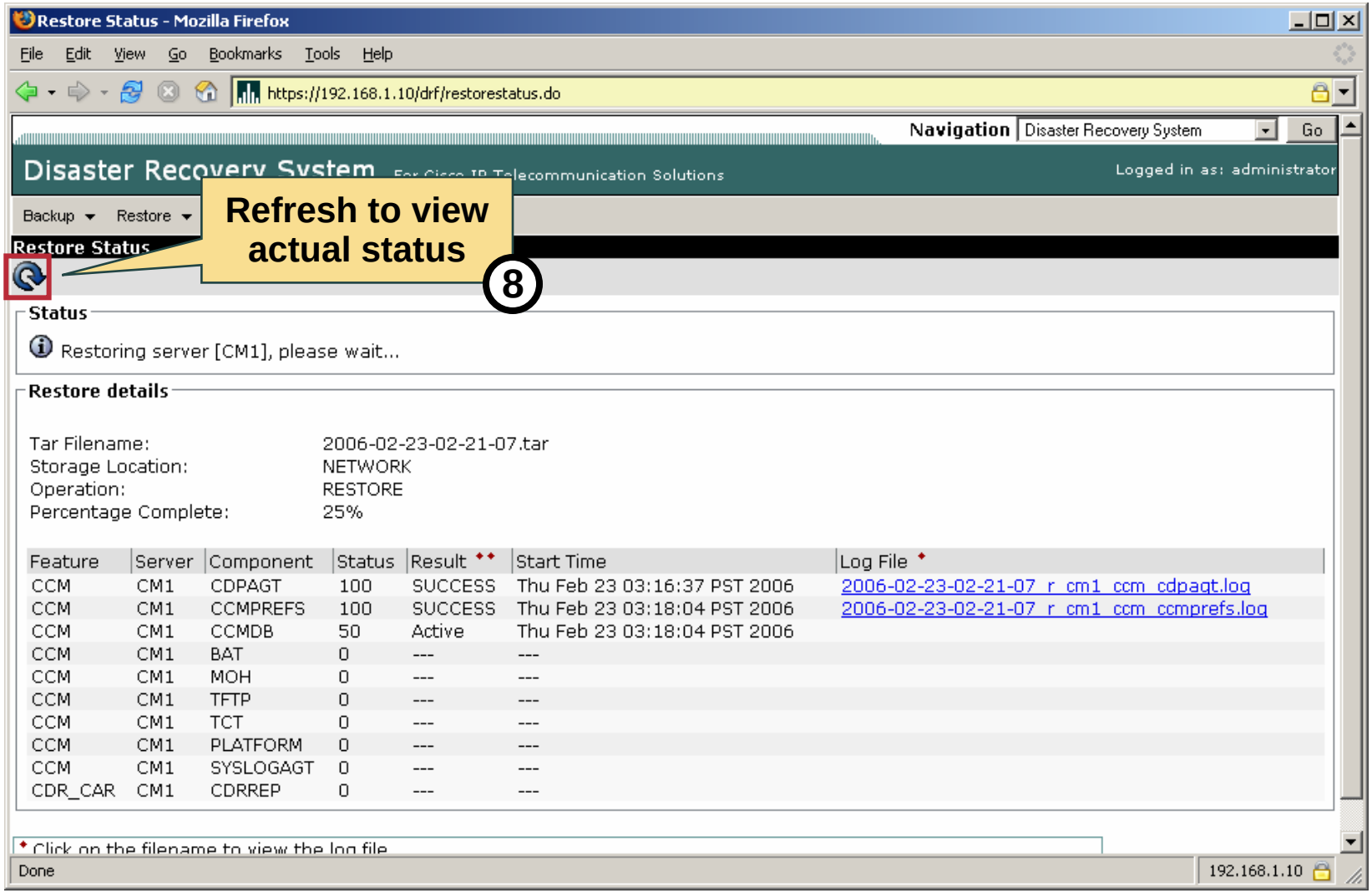
Backup and Restore Procedure: Restore Wizard



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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
```

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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

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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

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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.

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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.

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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.

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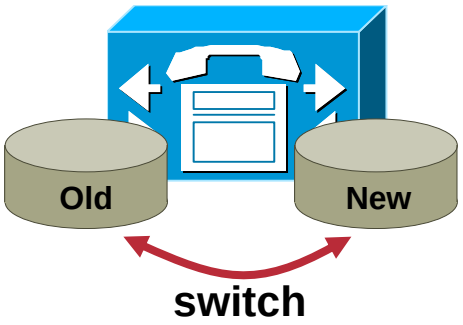
Dual Partitions and Switch Version

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.**



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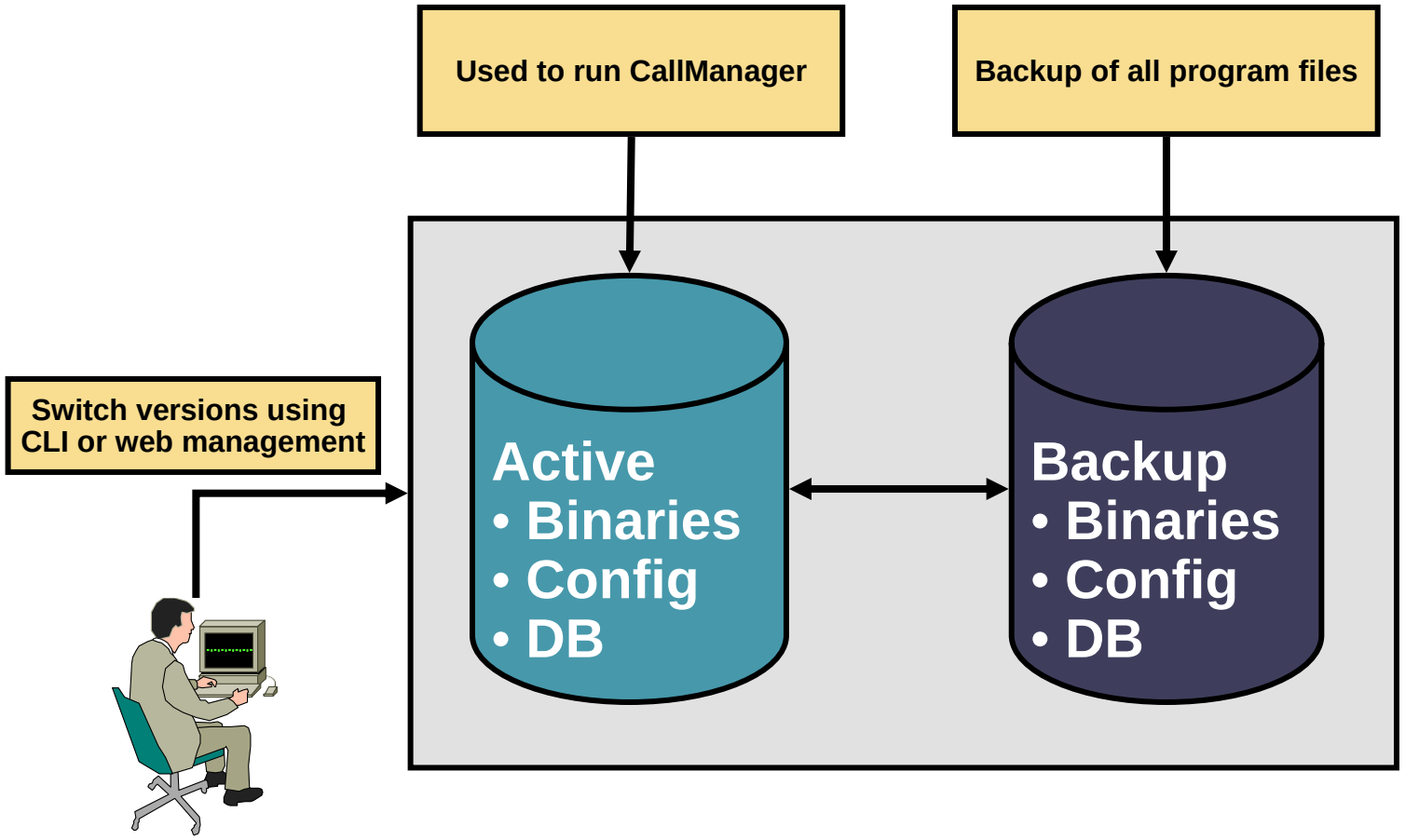
Dual Partitions (Switch Version)



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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.

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Deploying Cisco Unified CallManager 5.0 Endpoints

Configuring Catalyst Switches for Endpoints

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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

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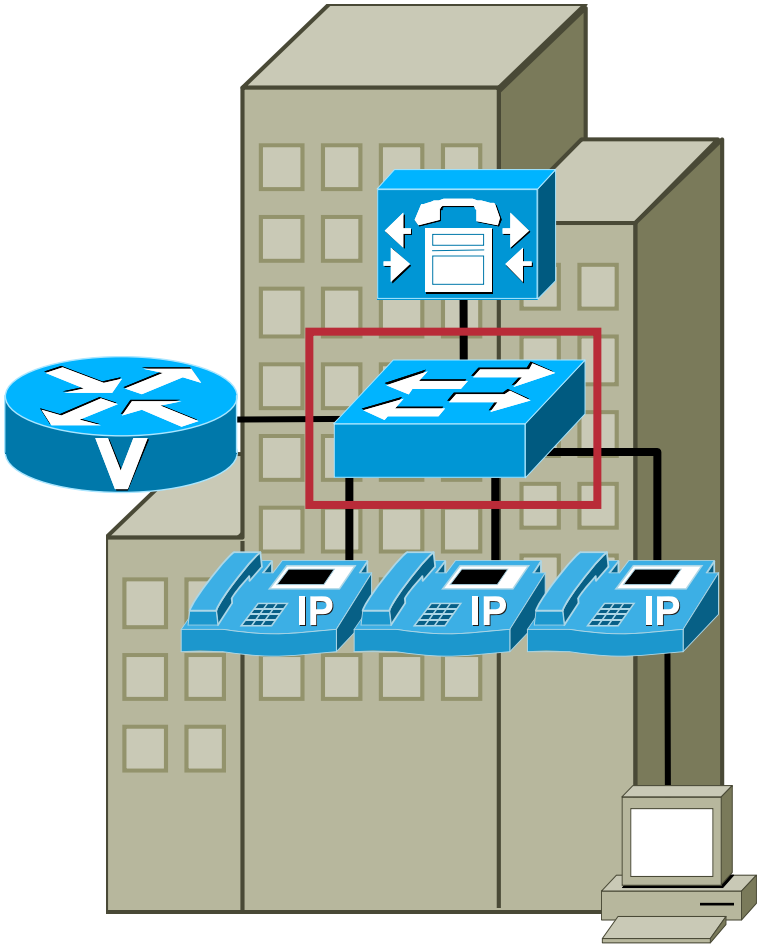
Catalyst Switch Role in IP Telephony

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



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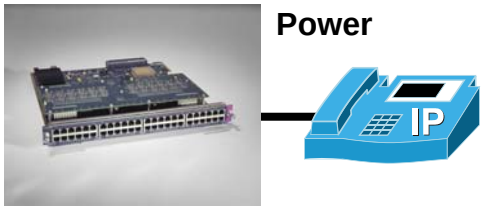
Powering the Cisco IP Phone

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)

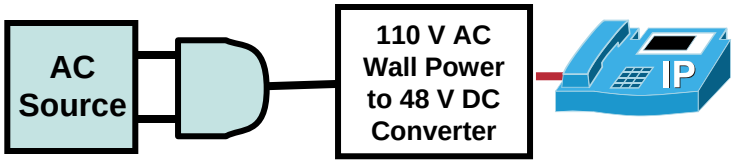


Midspan power injection:

- 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



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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.

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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.

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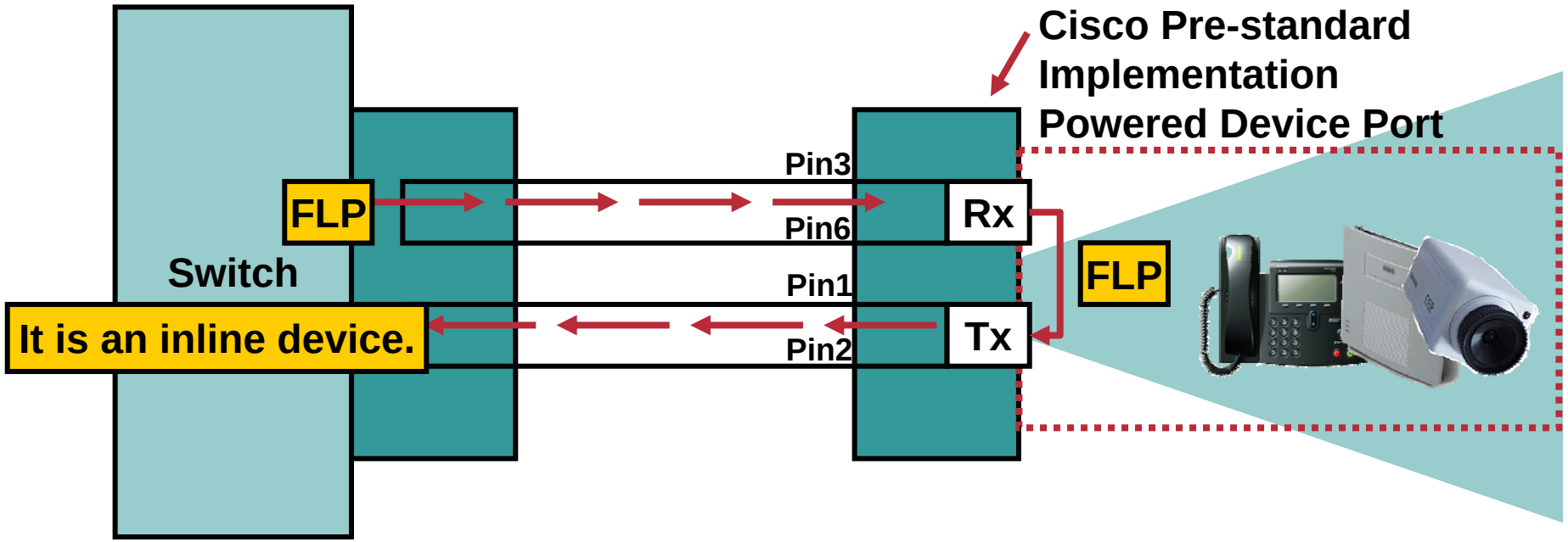
Types of PoE Delivery

Cisco pre-standard device detection:



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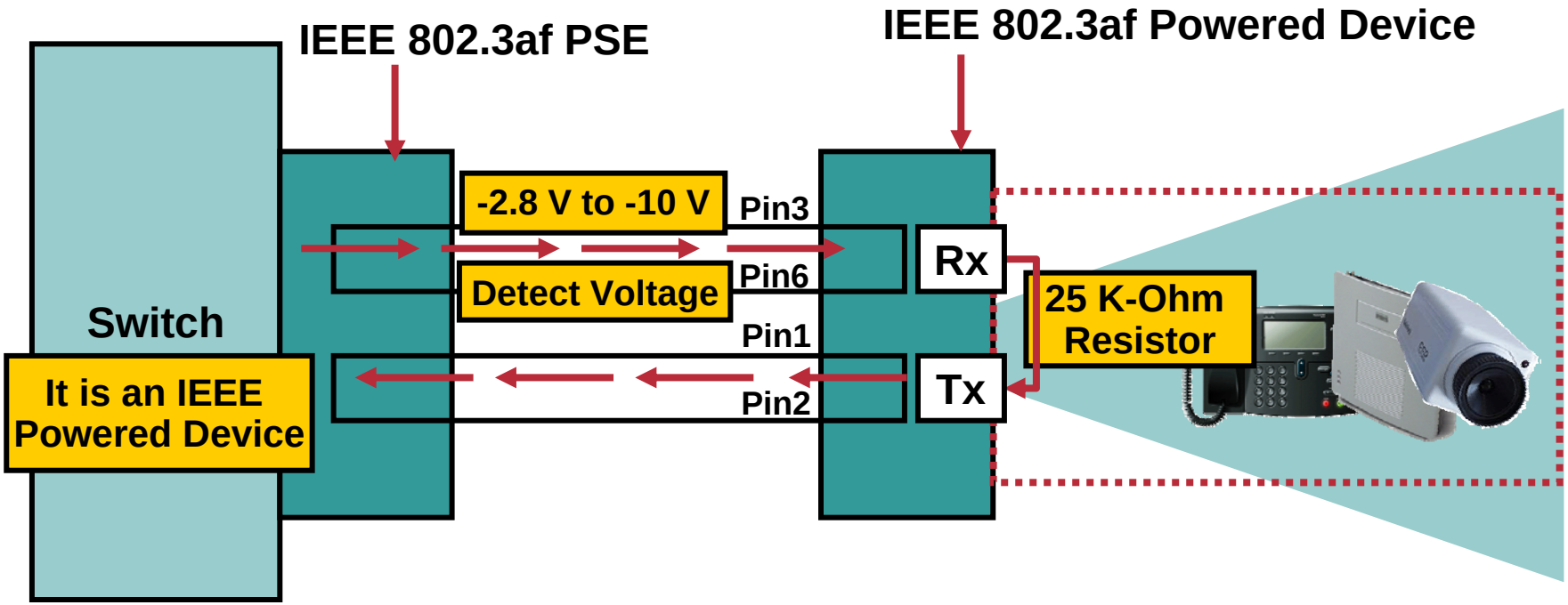
Types of PoE Delivery

IEEE 802.3af device detection:



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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**

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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
```

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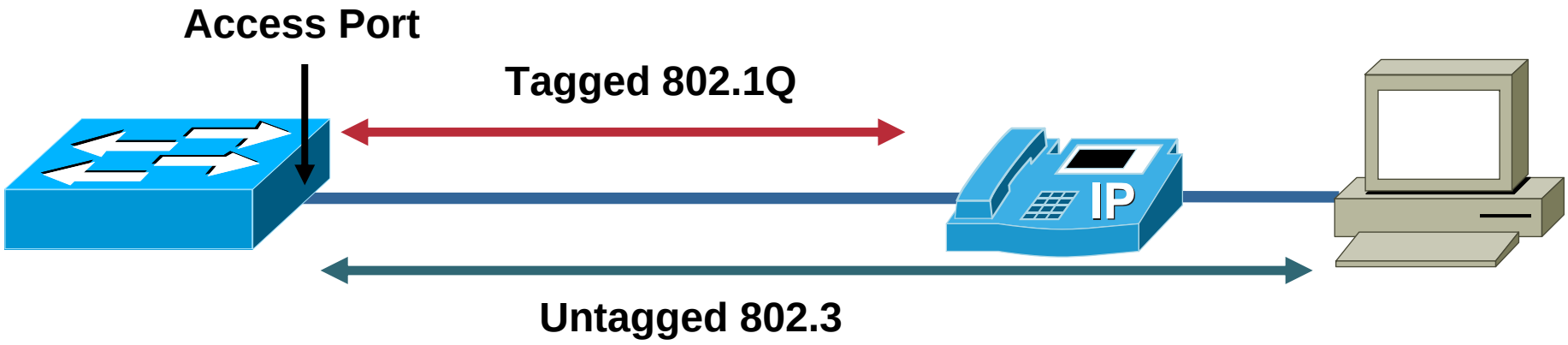
Configuring Dual VLANs

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.



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Configuring Dual VLANs

Configuring voice VLANs on native Cisco IOS using Catalyist 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)
```

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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
...
```

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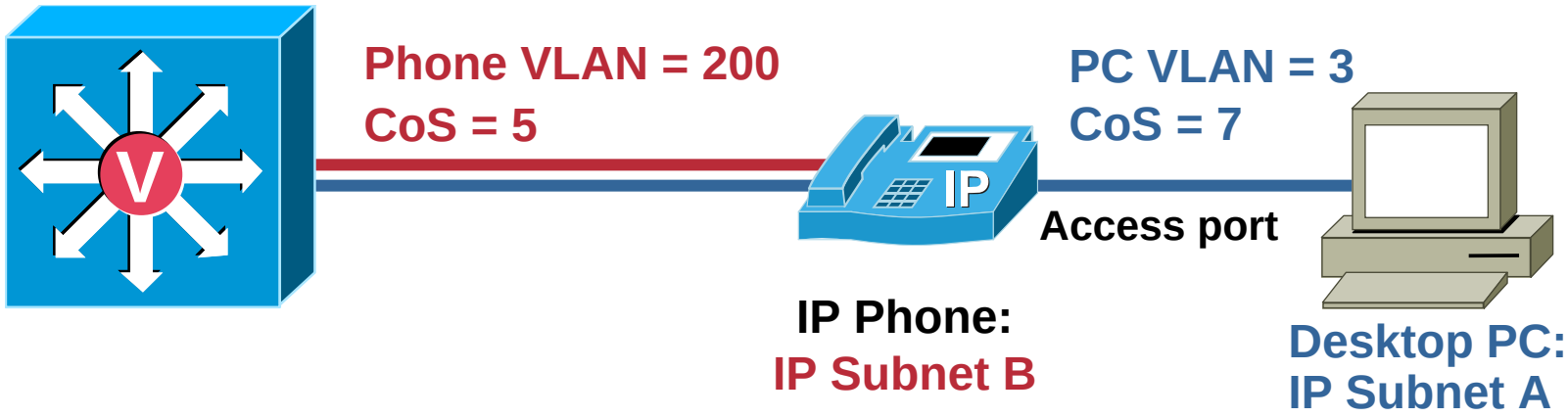
Configuring Class of Service

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

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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.**

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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.

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Deploying Cisco Unified CallManager 5.0 Endpoints

Configuring SCCP Endpoints in Cisco Unified CallManager 5.0

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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

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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.**

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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.**

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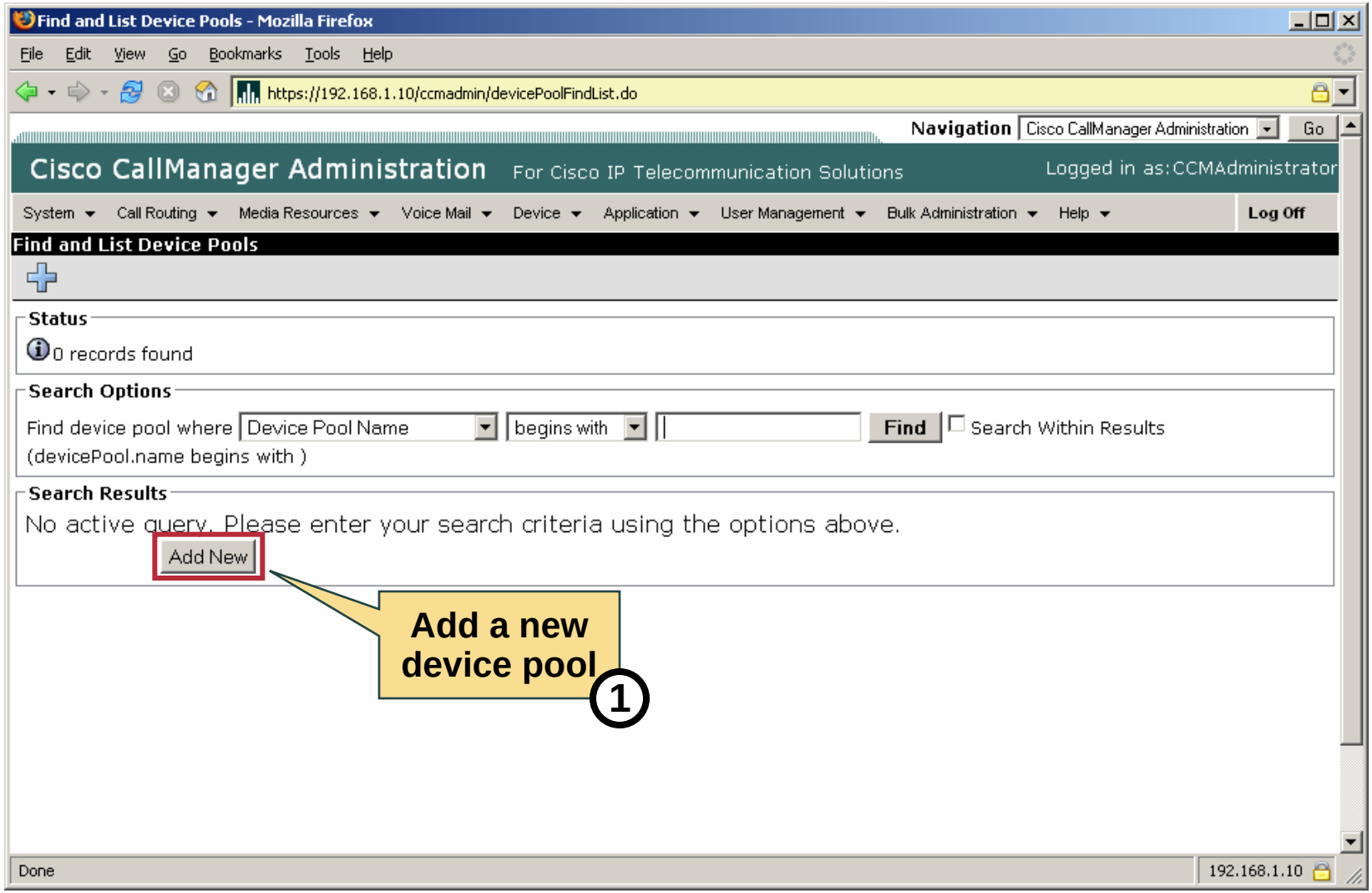
Configuring Device Pools



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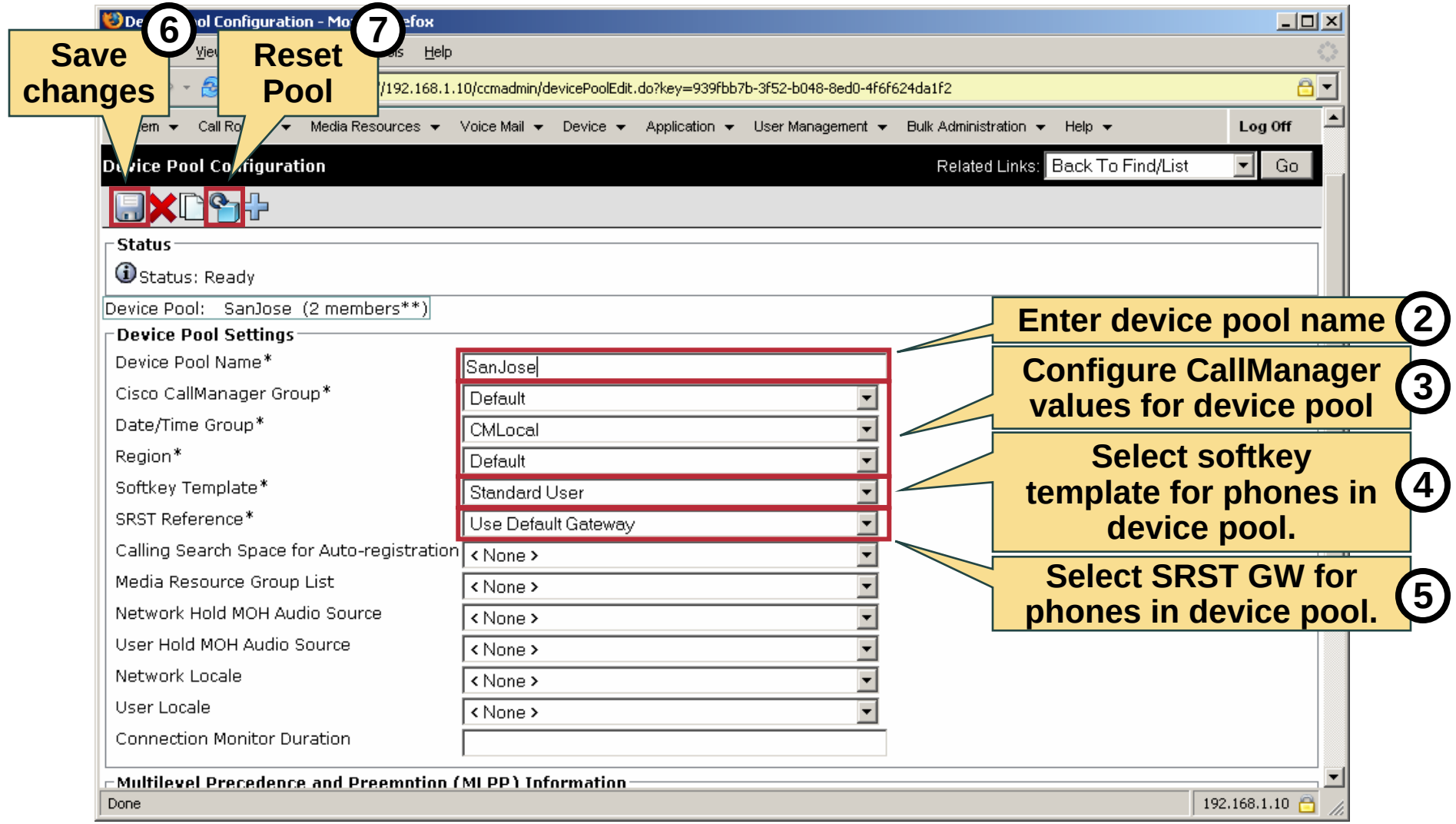
Configuring Device Pools



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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.

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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.

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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.**

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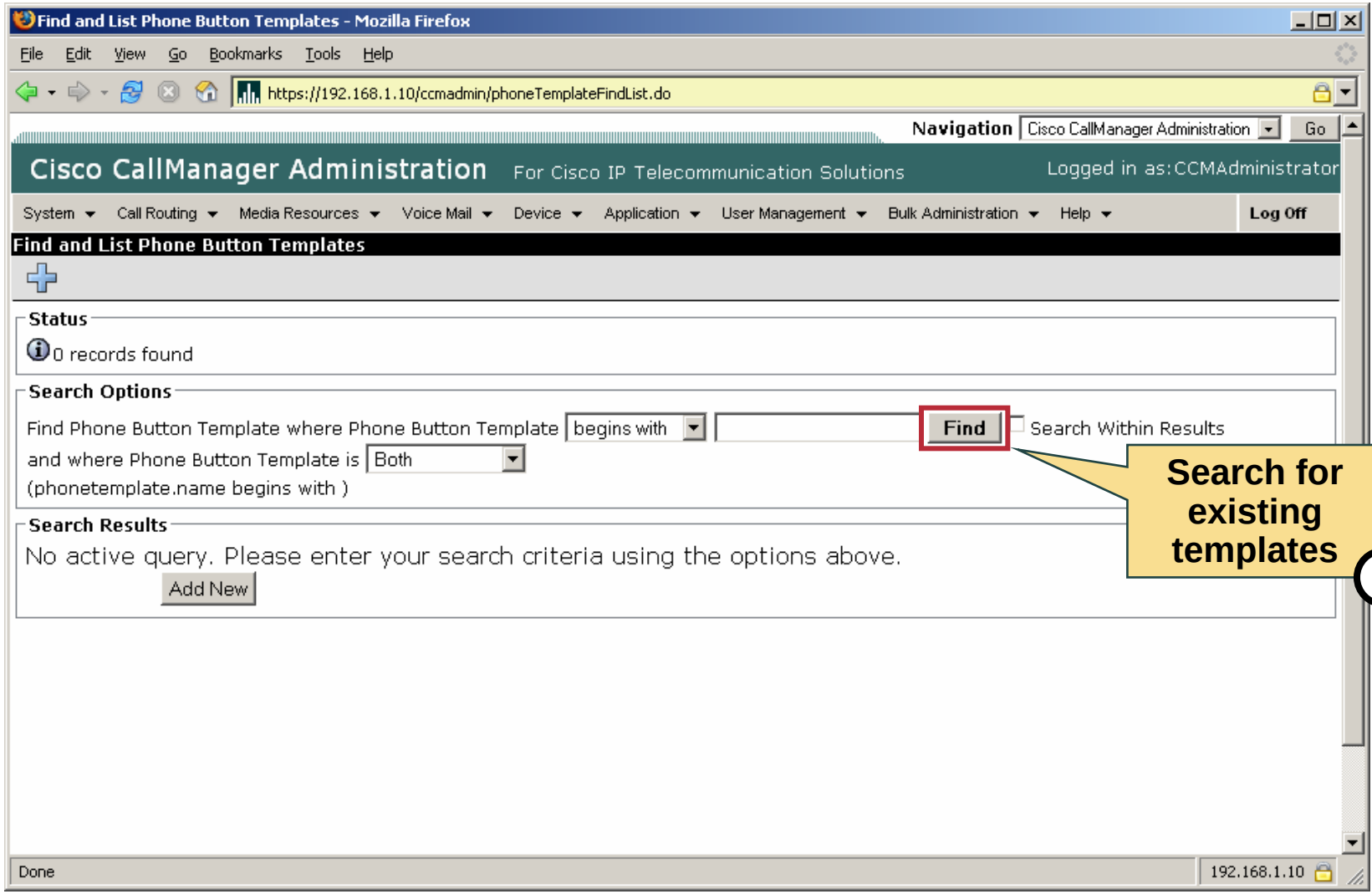
IP Phone Button Templates



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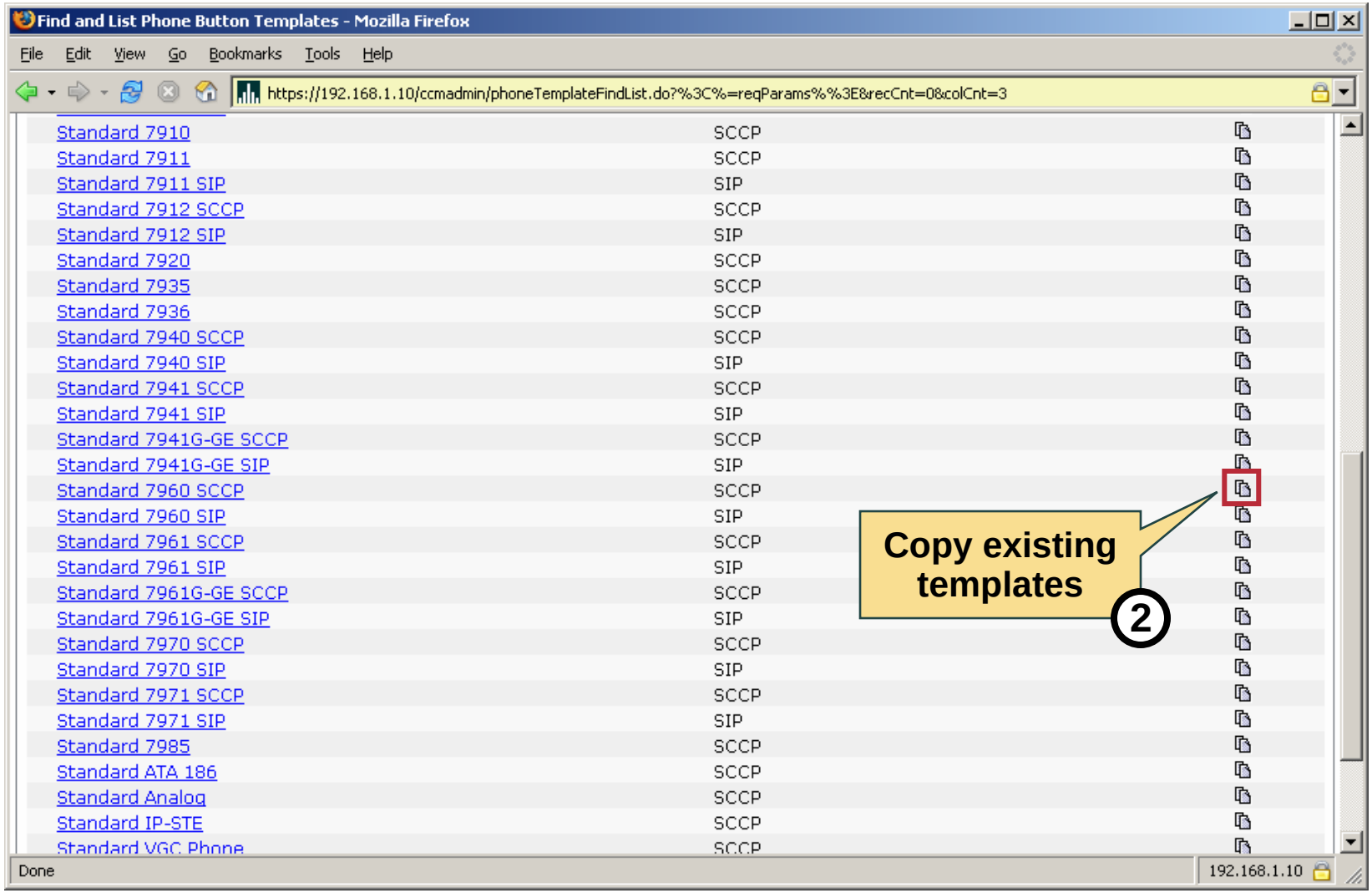
IP Phone Button Templates



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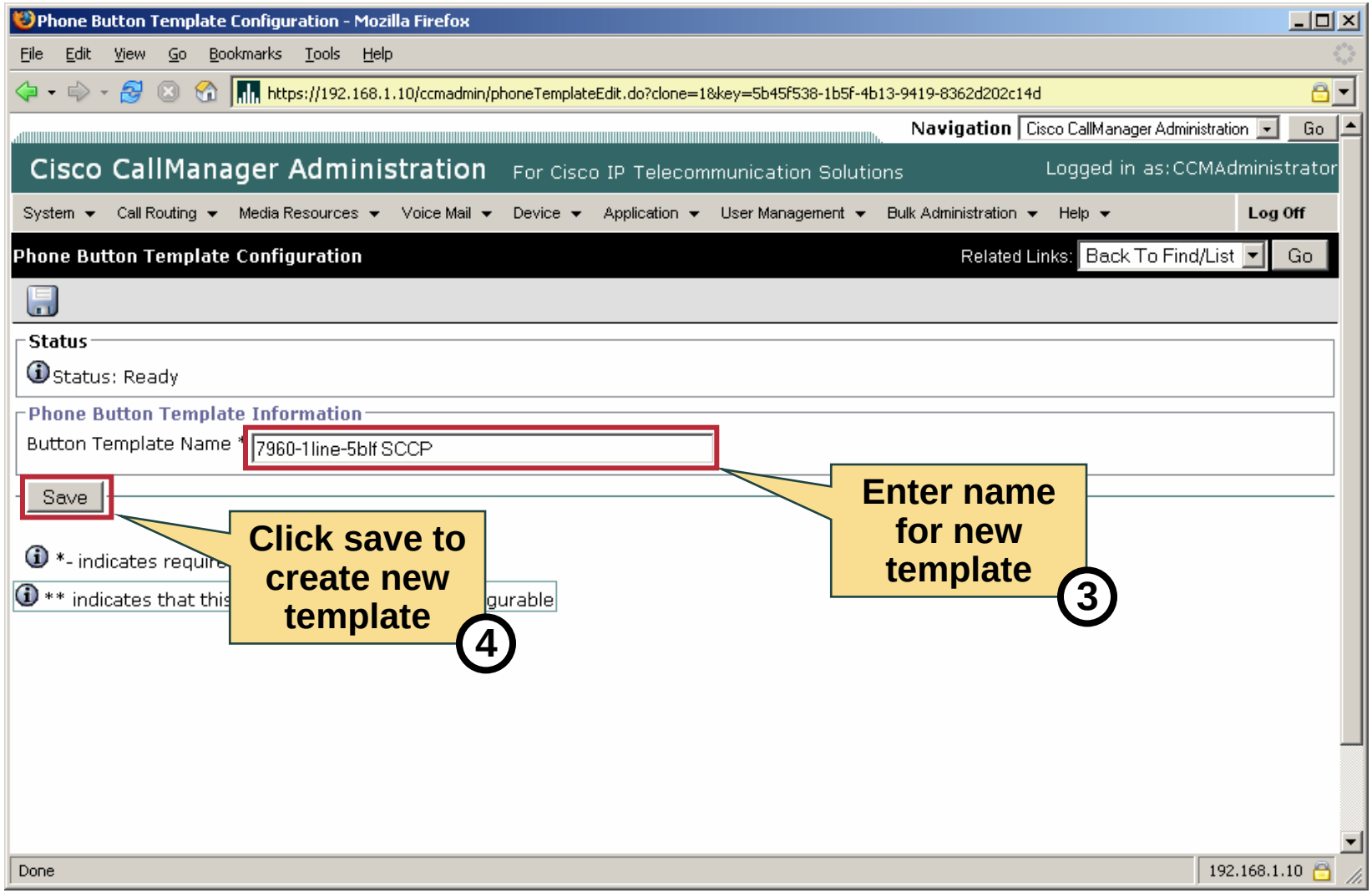
IP Phone Button Templates



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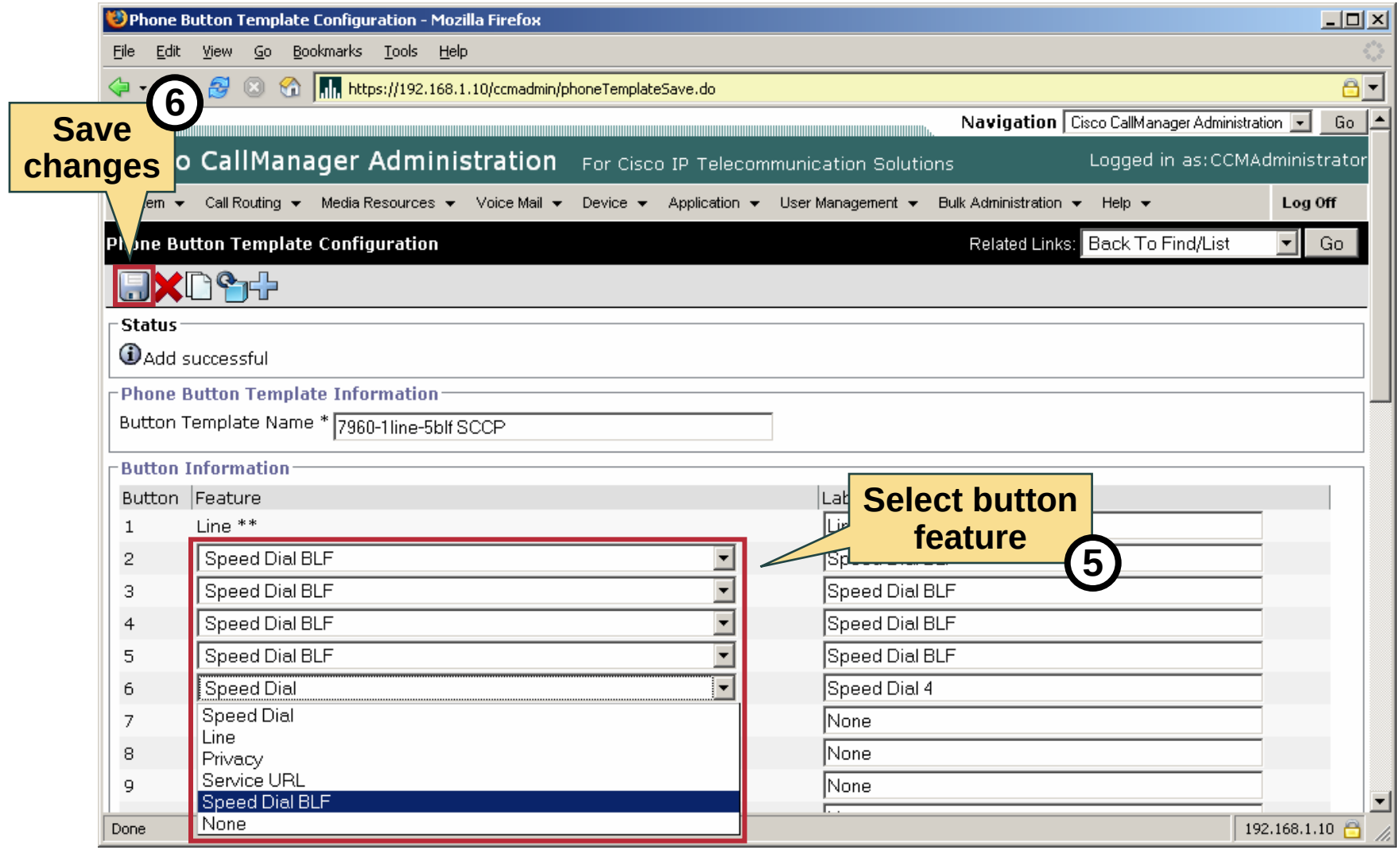
IP Phone Button Templates



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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.**

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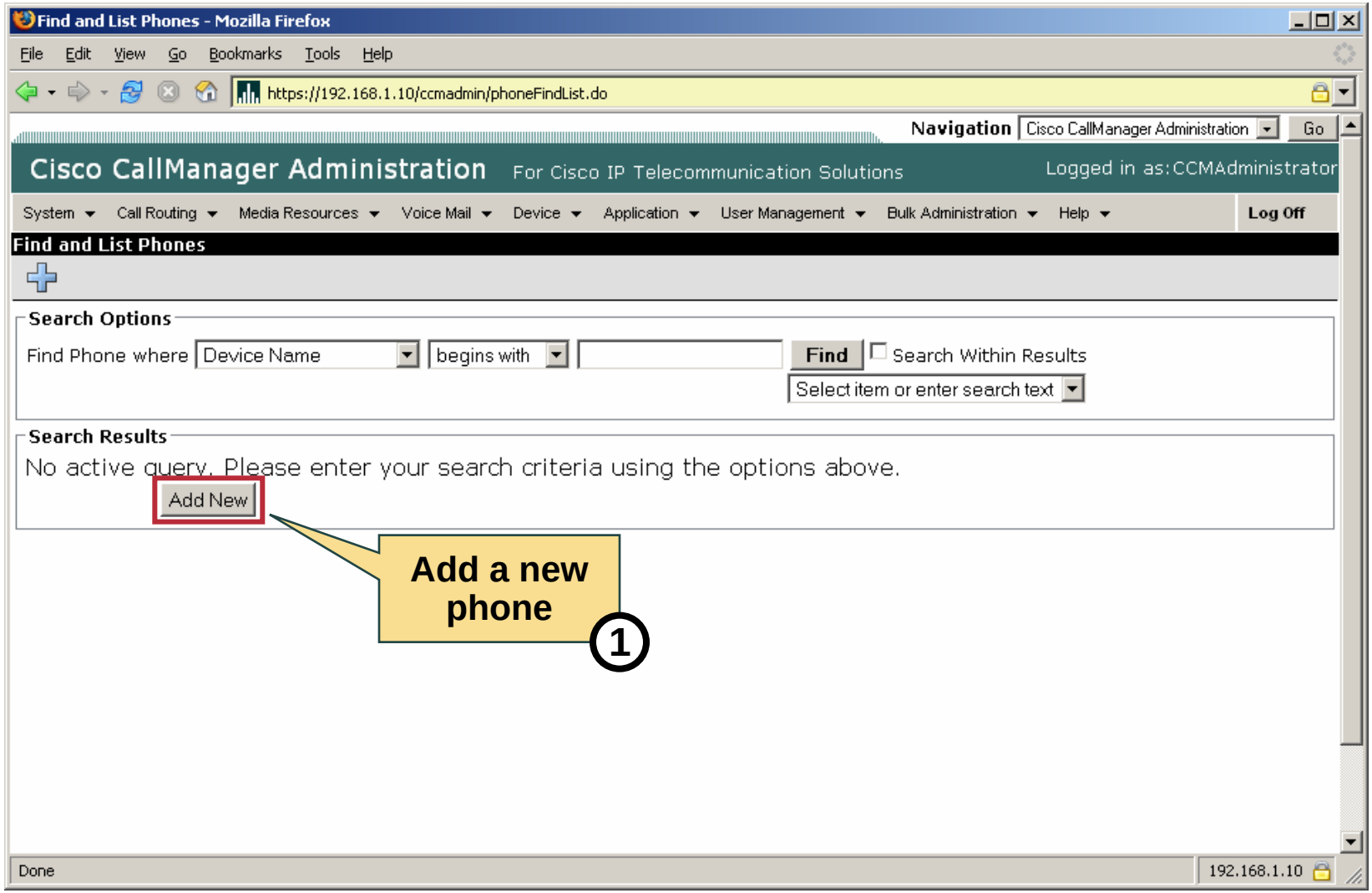
Manual IP Phone Configuration



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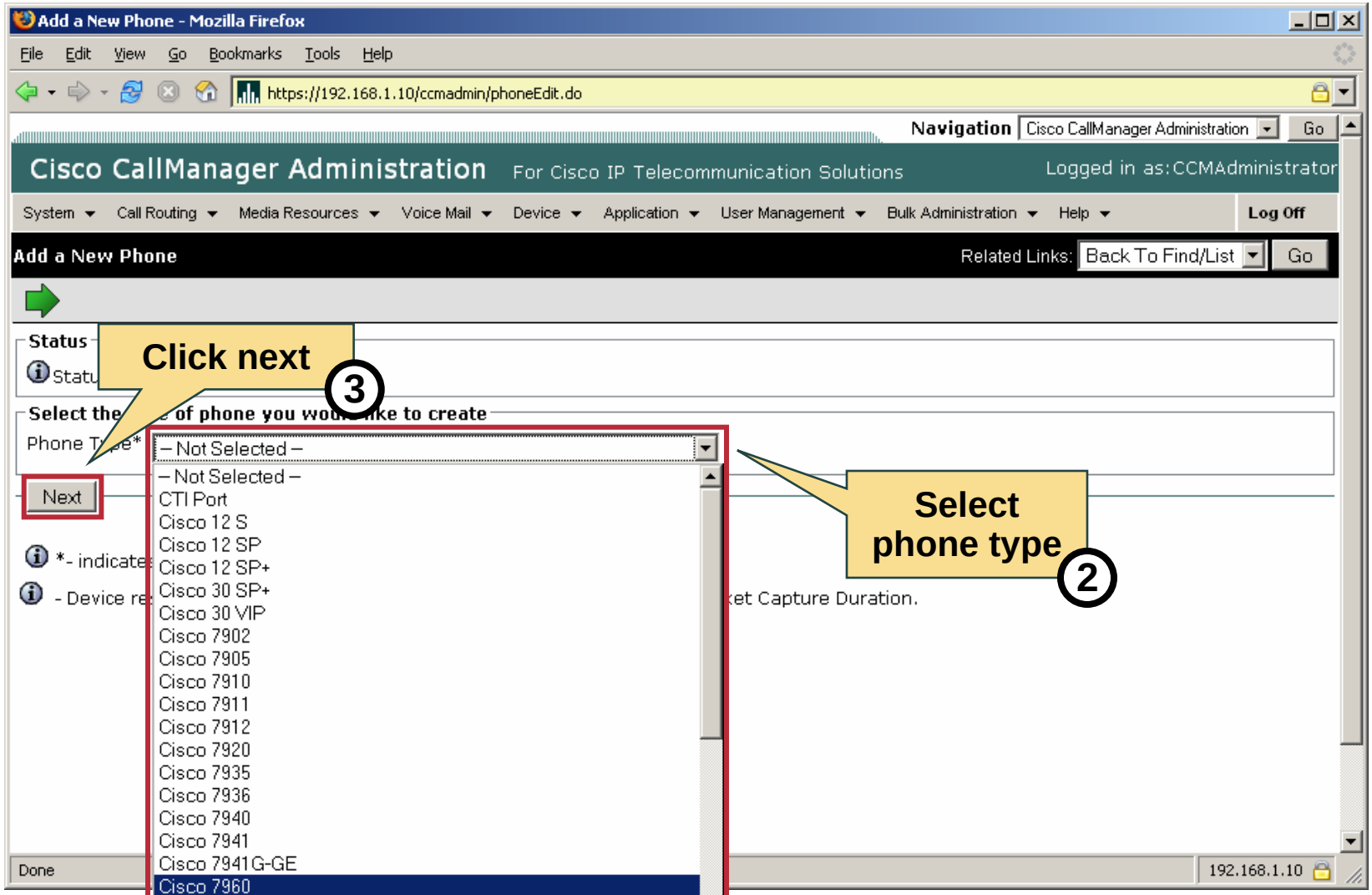
Manual IP Phone Configuration



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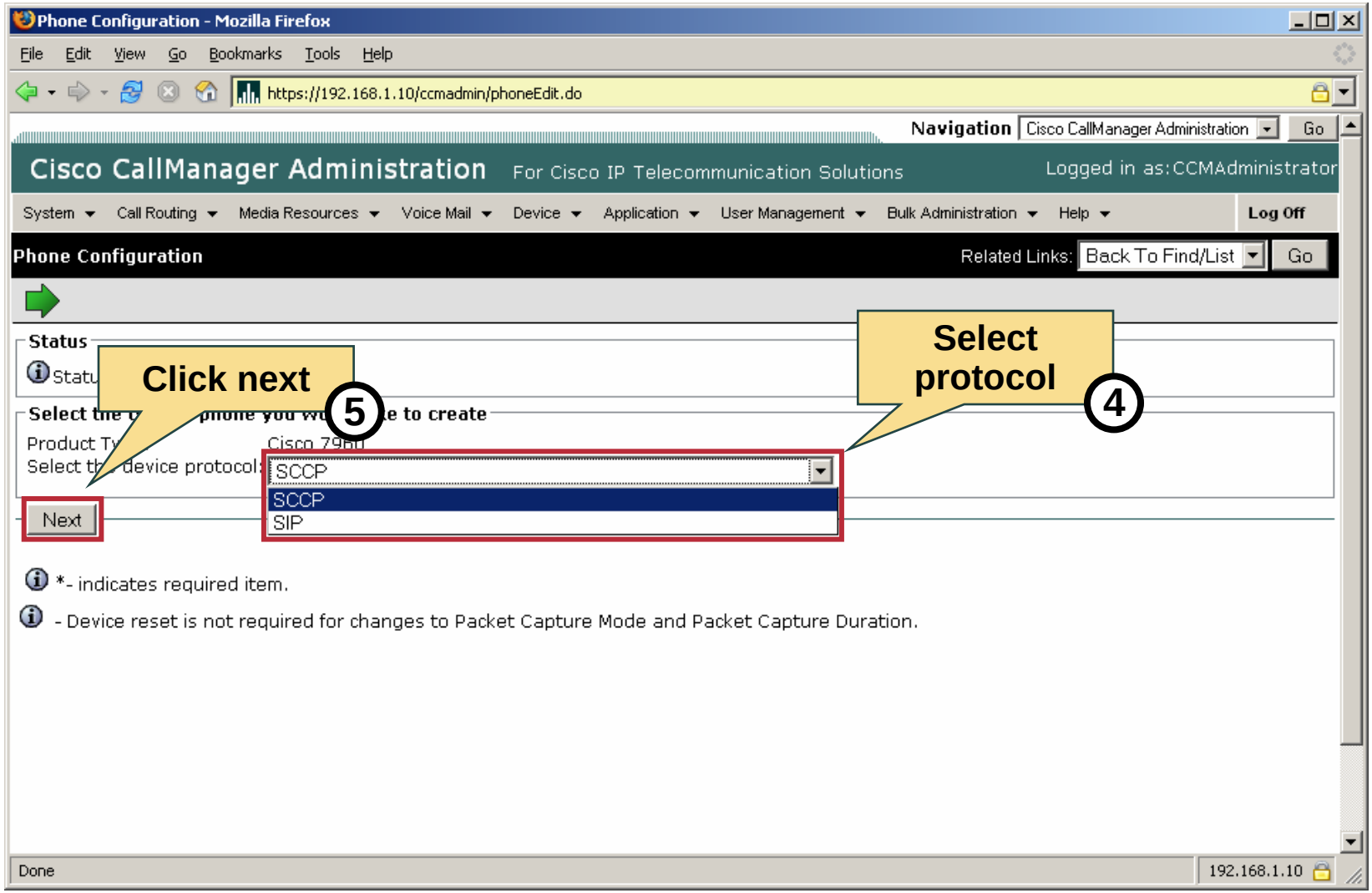
Manual IP Phone Configuration



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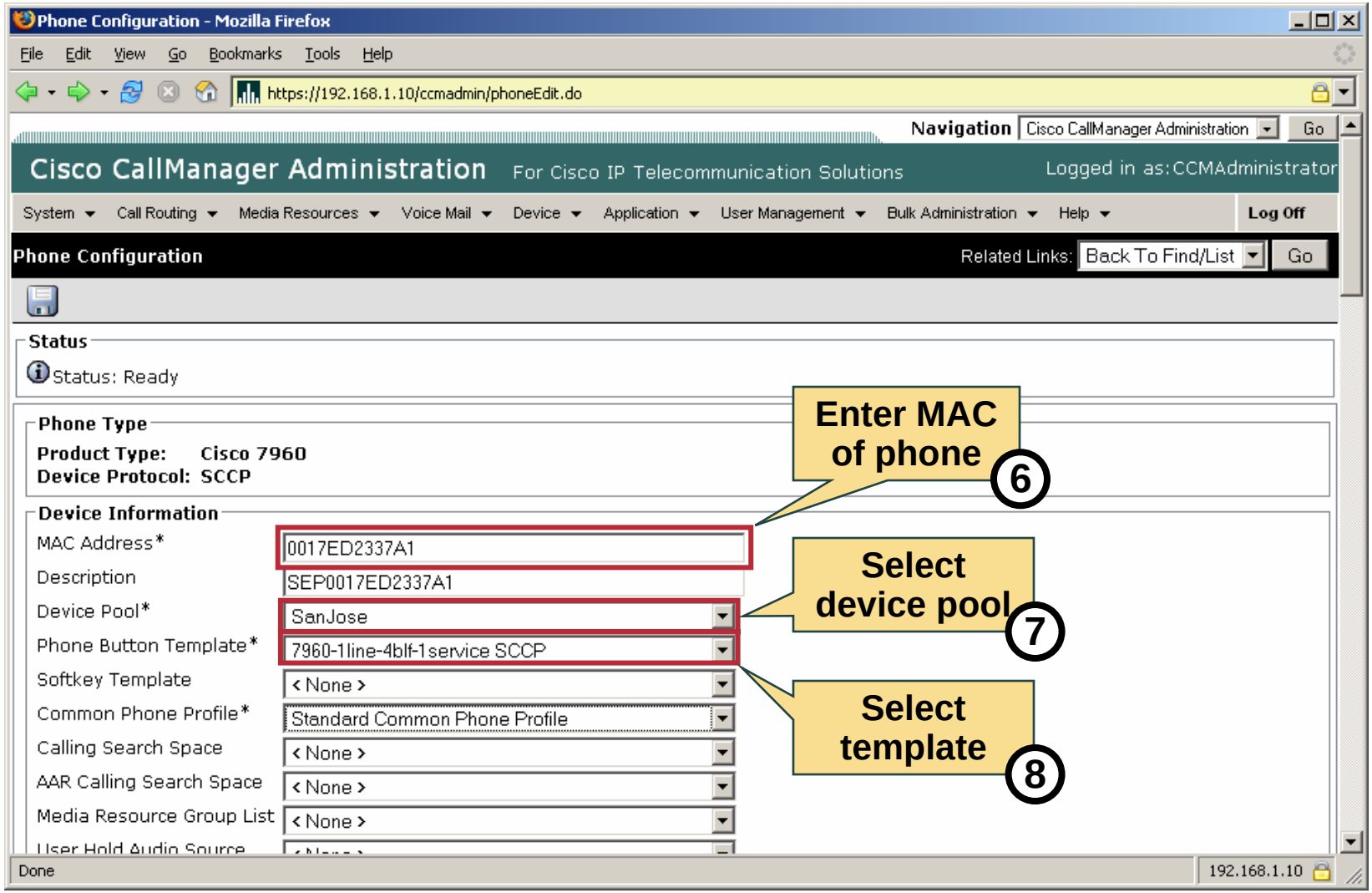
Manual IP Phone Configuration



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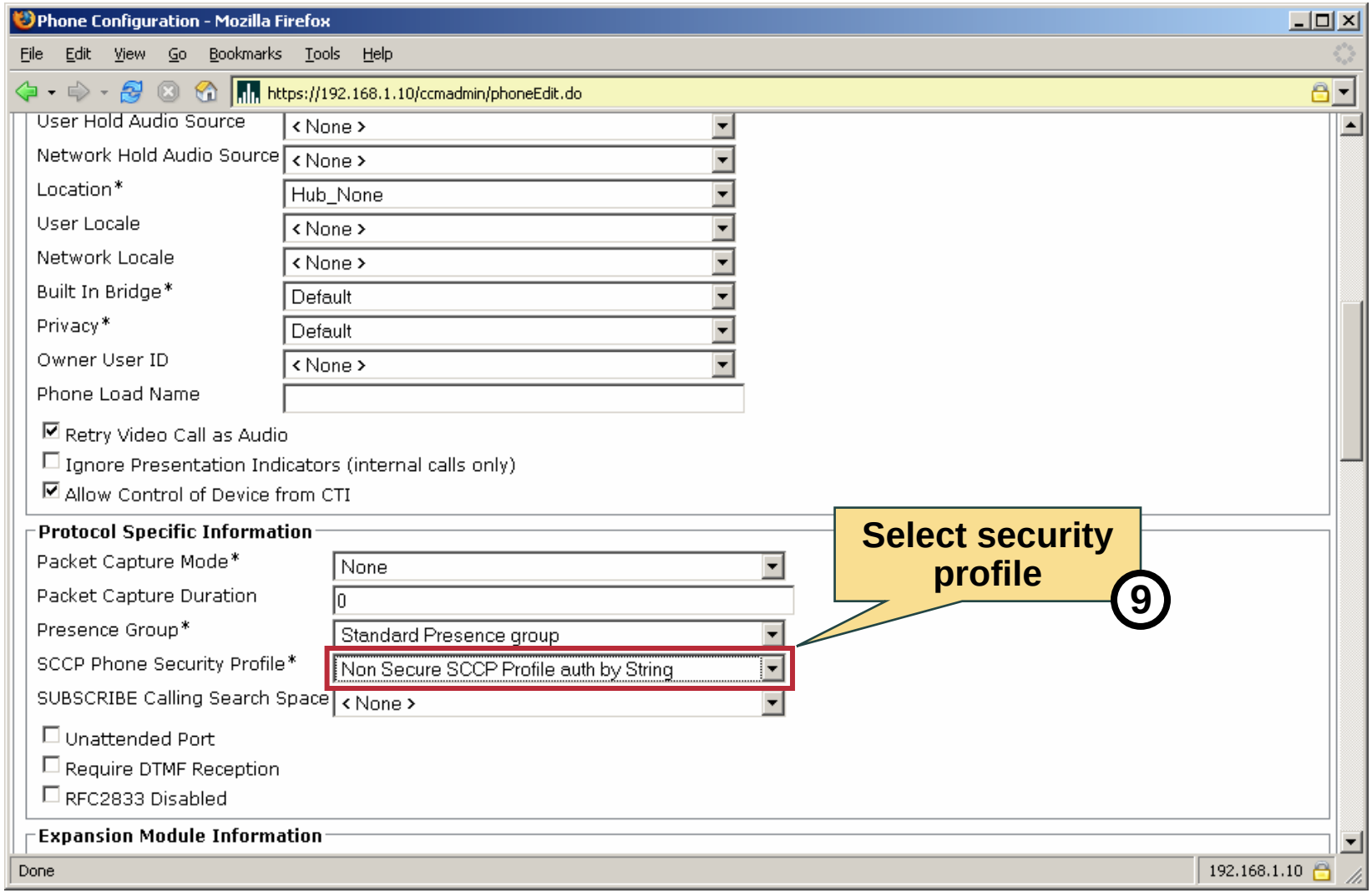
Manual IP Phone Configuration



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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.

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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.**

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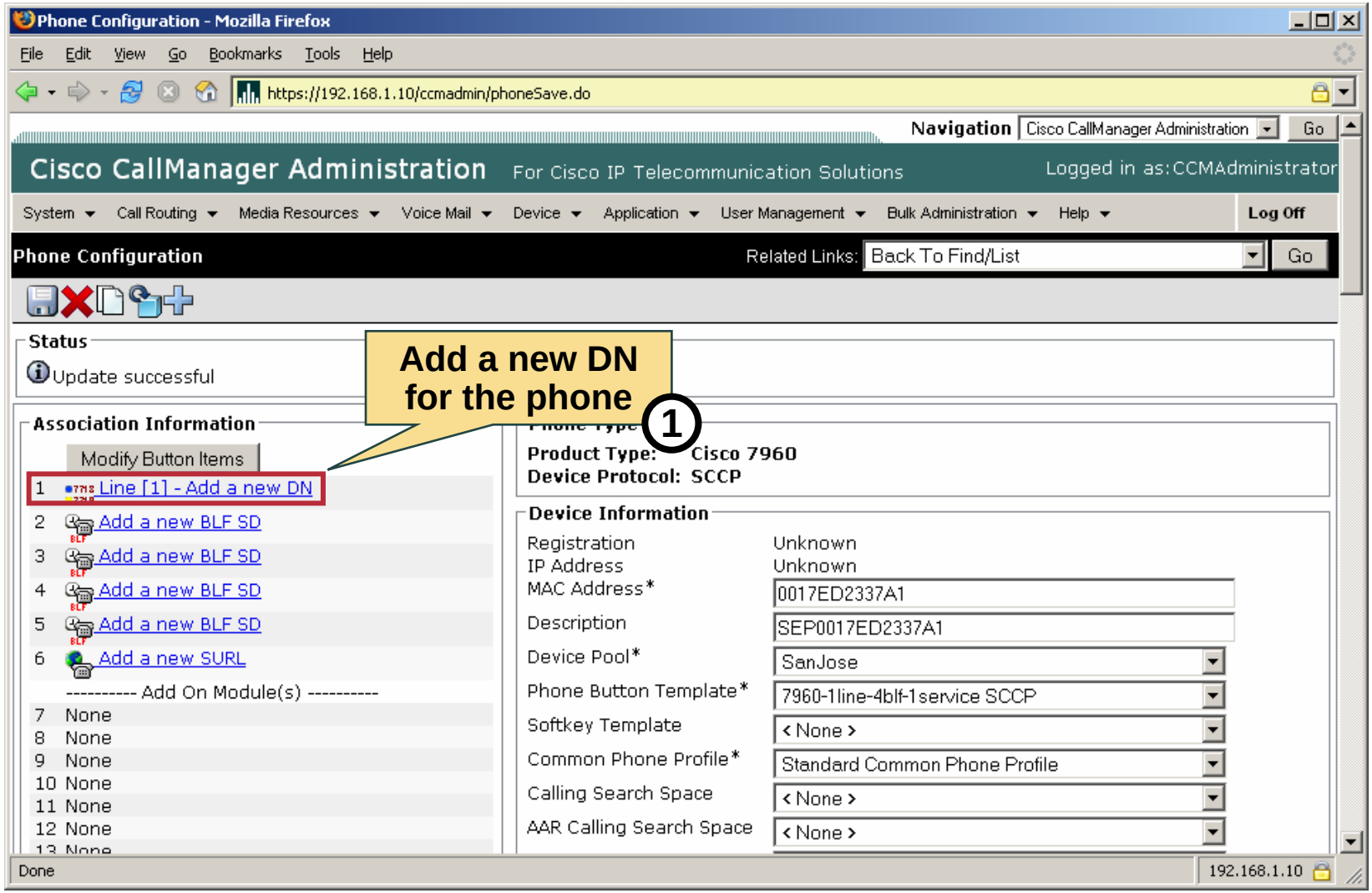
Line Configuration



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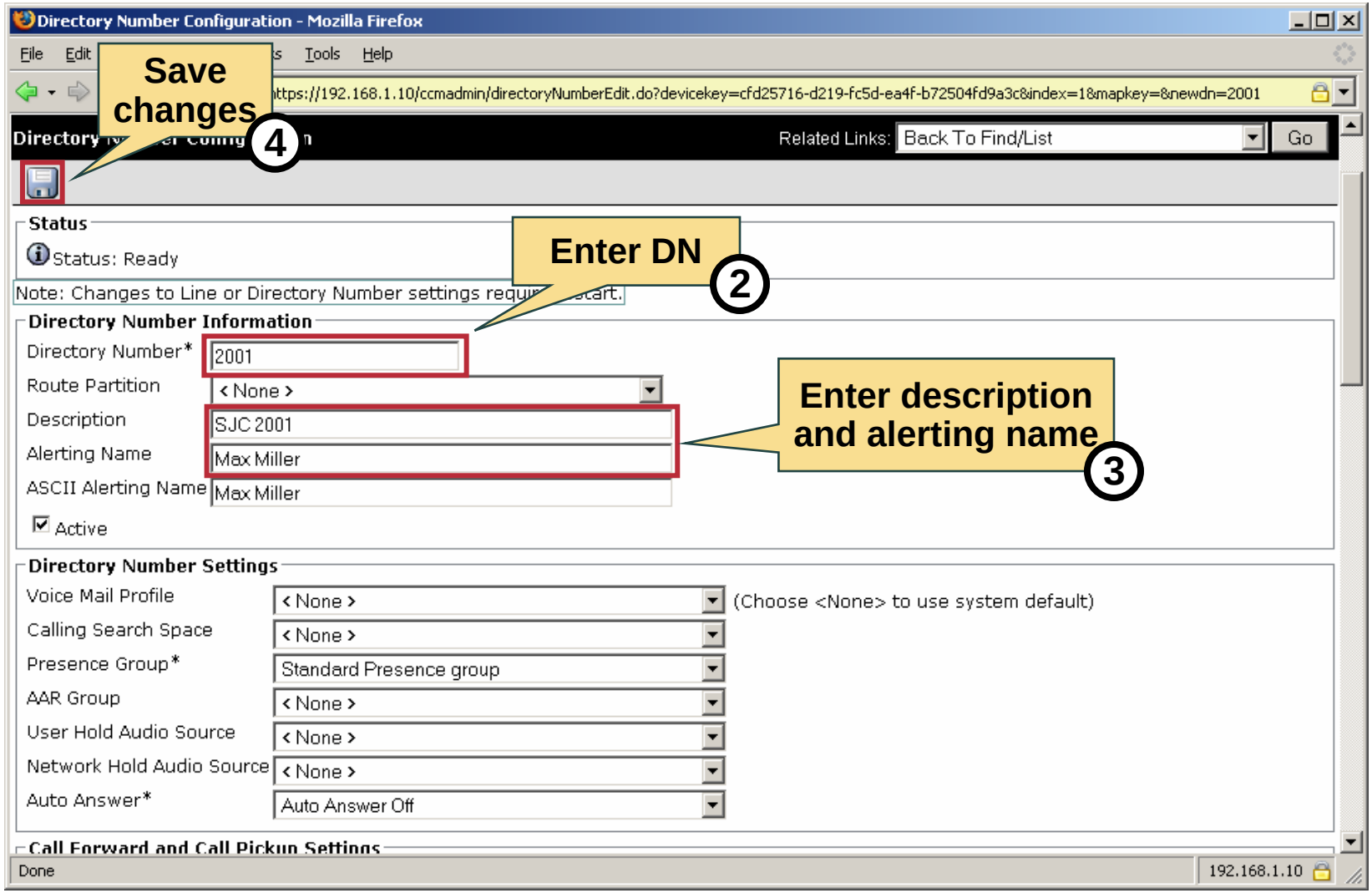
Line Configuration



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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.

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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.**

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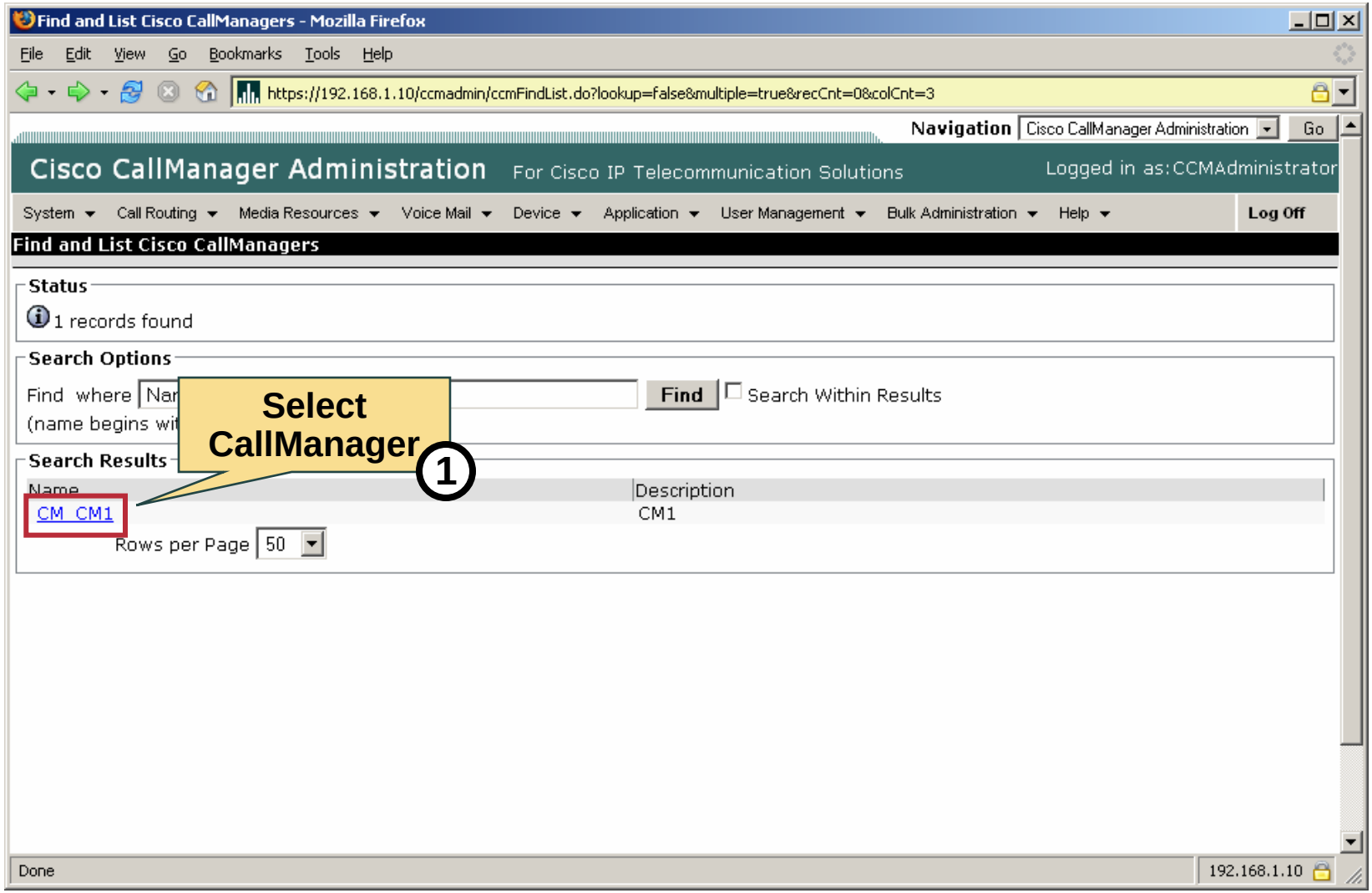
Configuring IP Phone Auto-Registration



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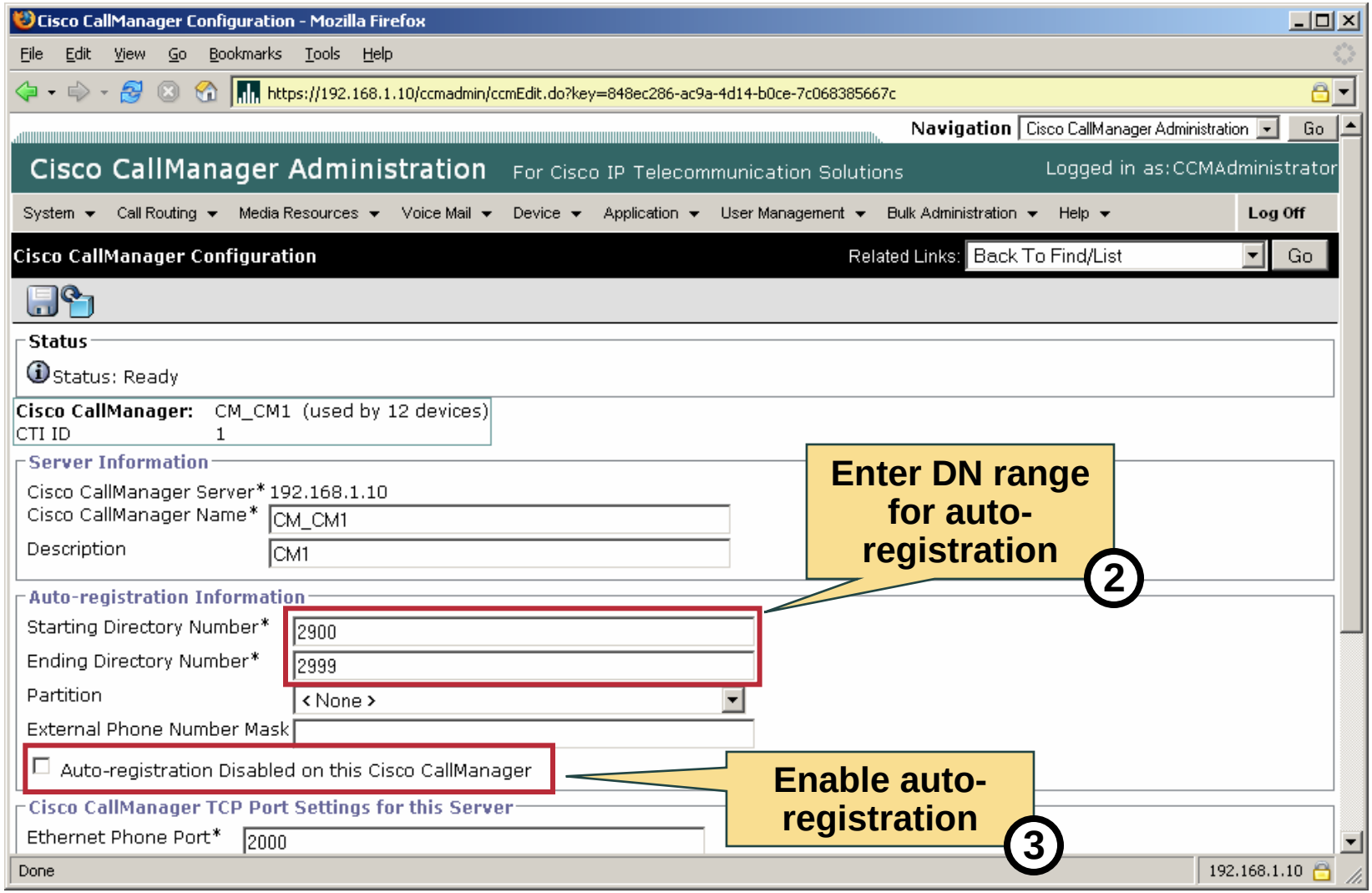
Configuring IP Phone Auto-Registration



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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).**

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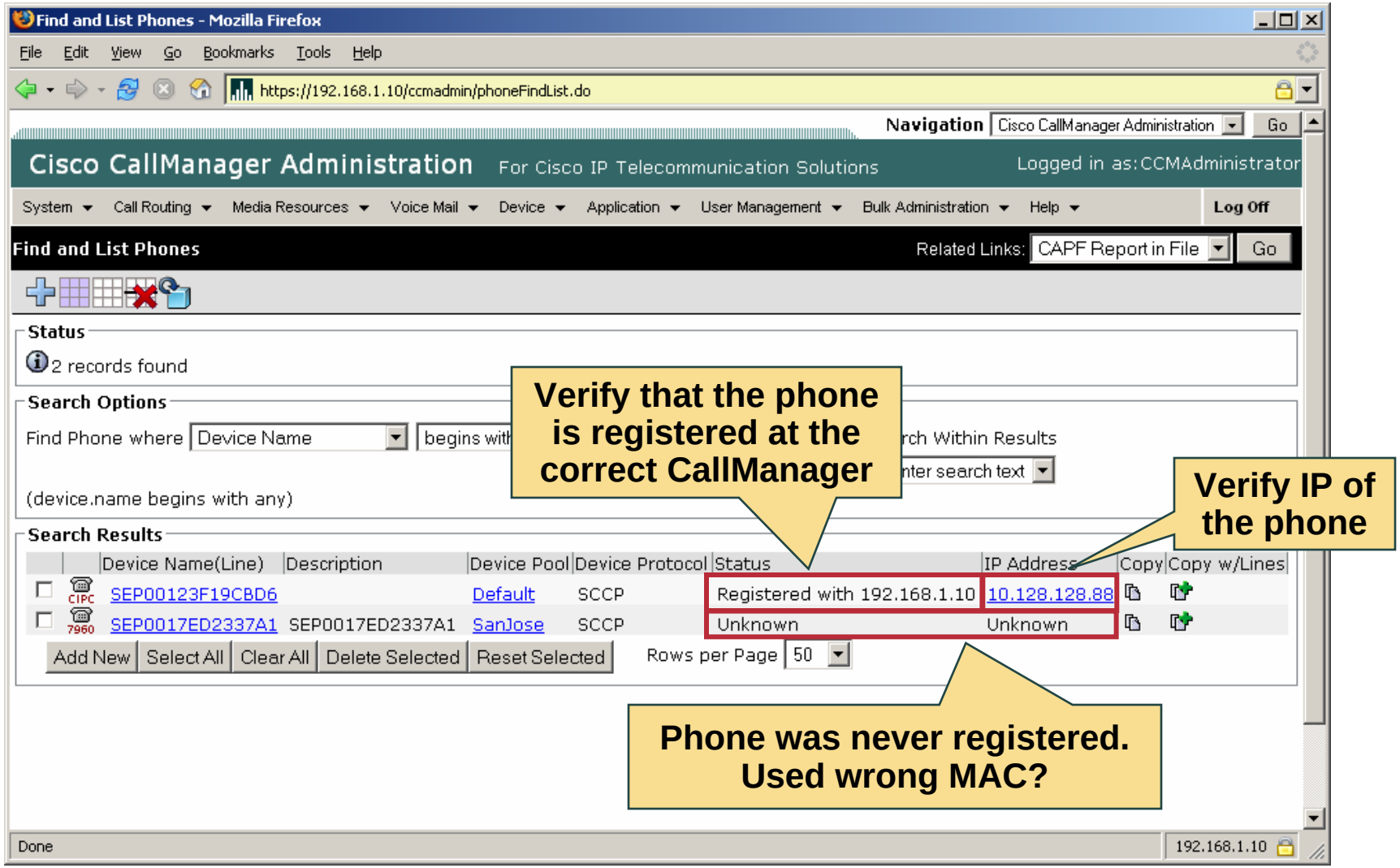
Verify Endpoint Configuration



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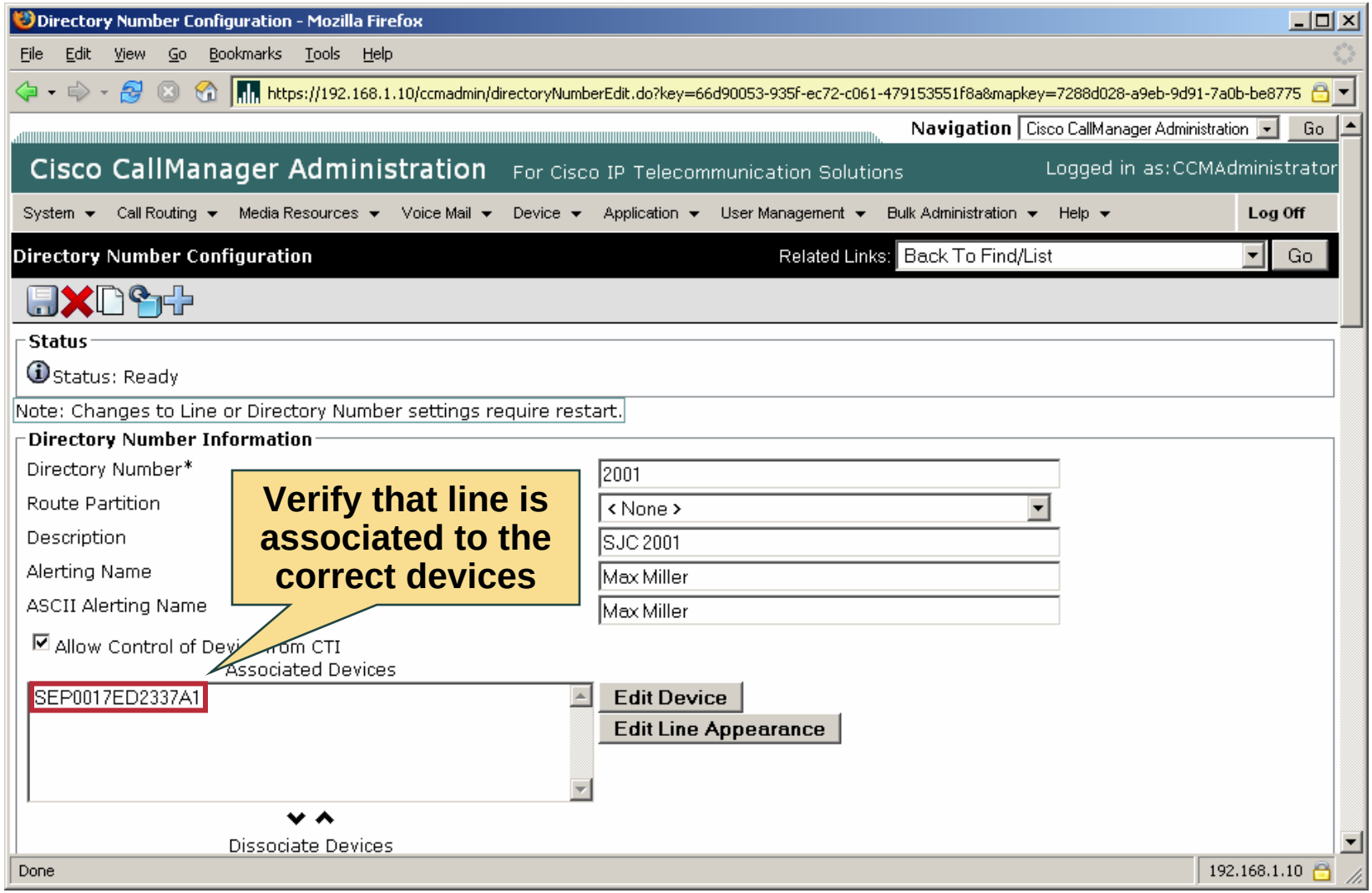
Verify Endpoint Configuration



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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.

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Deploying Cisco Unified CallManager 5.0 Endpoints

Configuring SIP Endpoints in Cisco Unified CallManager 5.0

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Objectives

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

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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.**

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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.**

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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.**

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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.**

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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

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SIP Phone Features

Phone to CCM interaction comparison:

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

[1] Part of the configuration file.
[2] In the future, 3rd-party phones may add this support. Failover can be supported today via DNS SRV.

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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.**

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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.**

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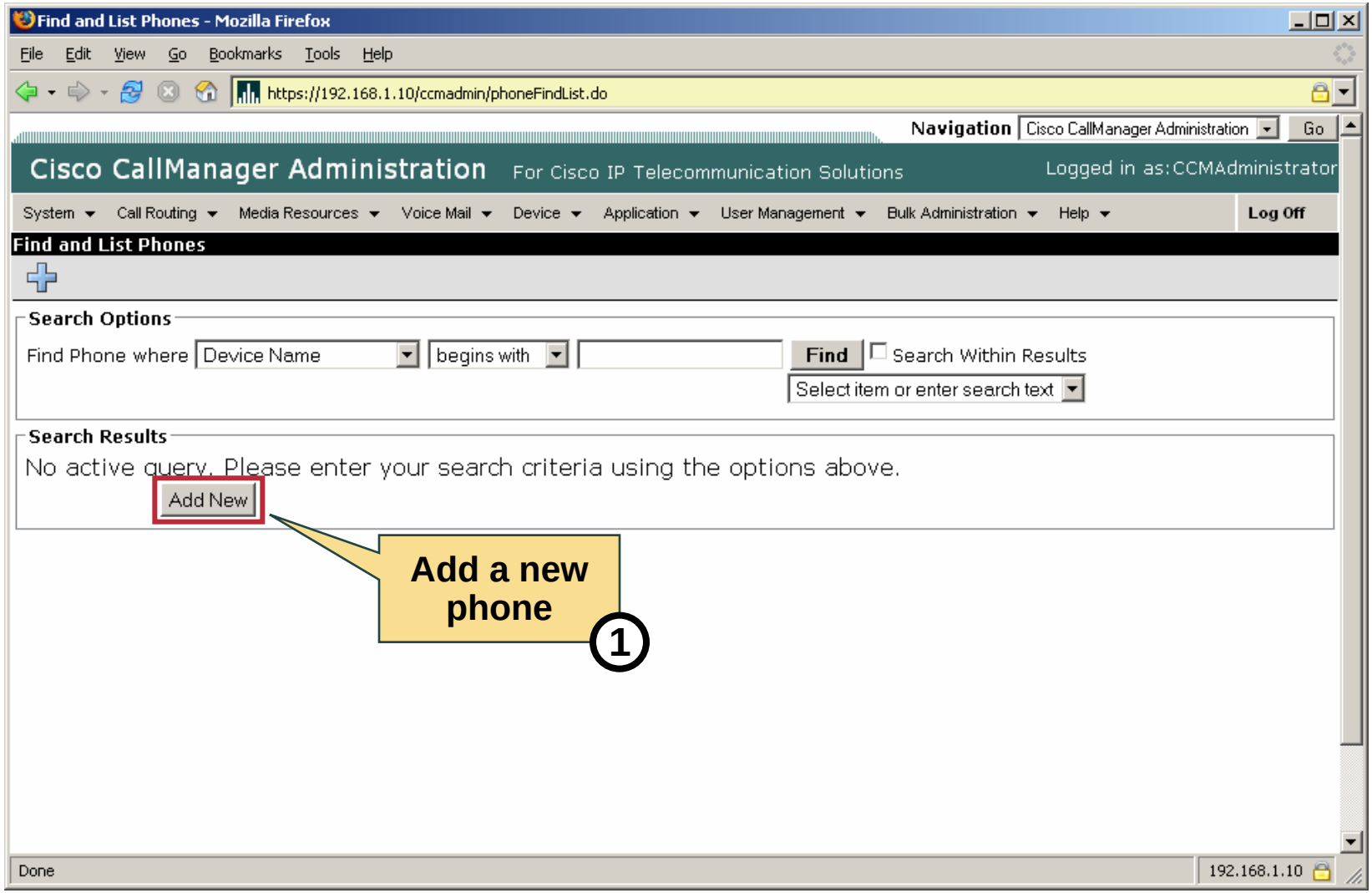
Configuring Cisco SIP Phones



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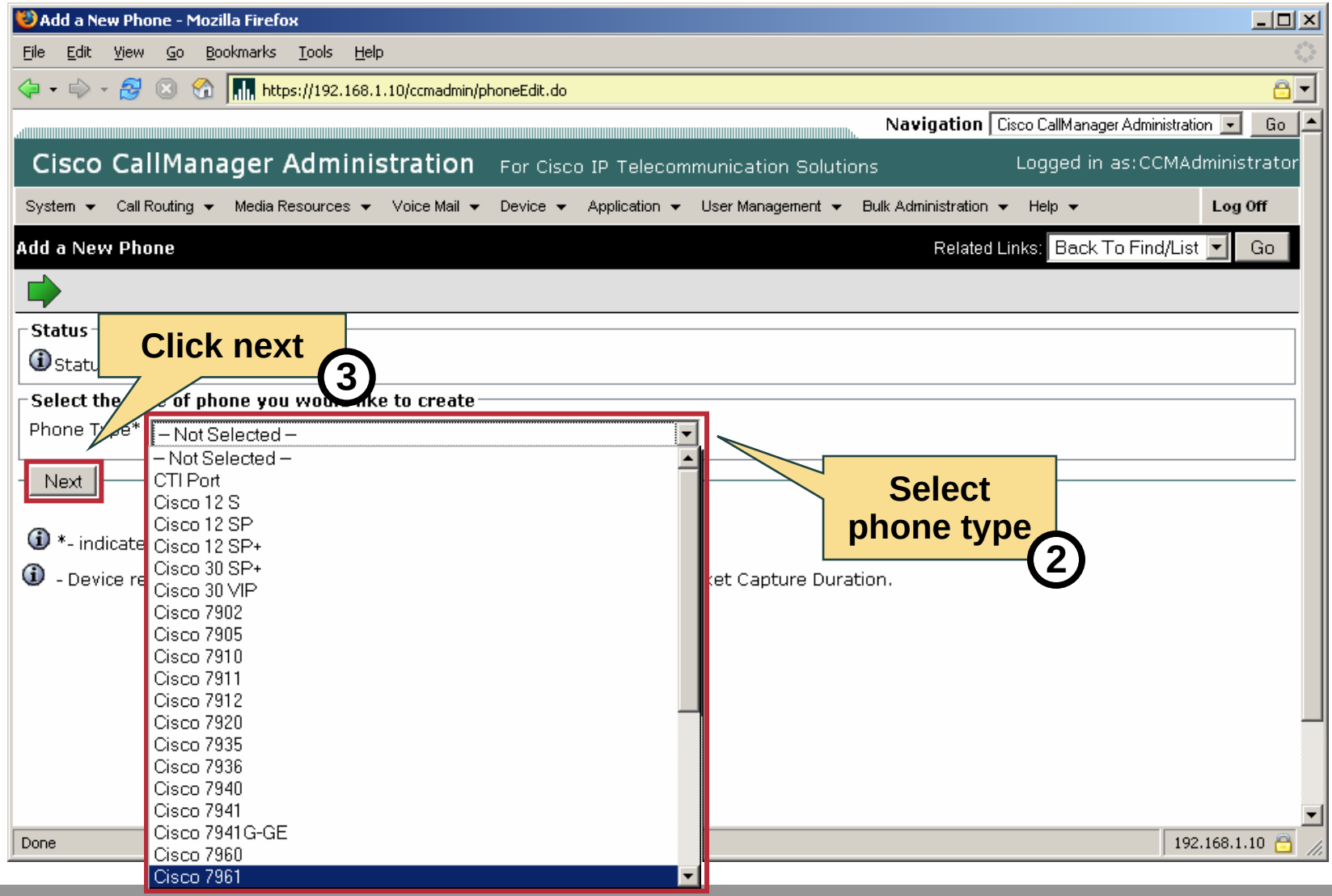
Configuring Cisco SIP Phones



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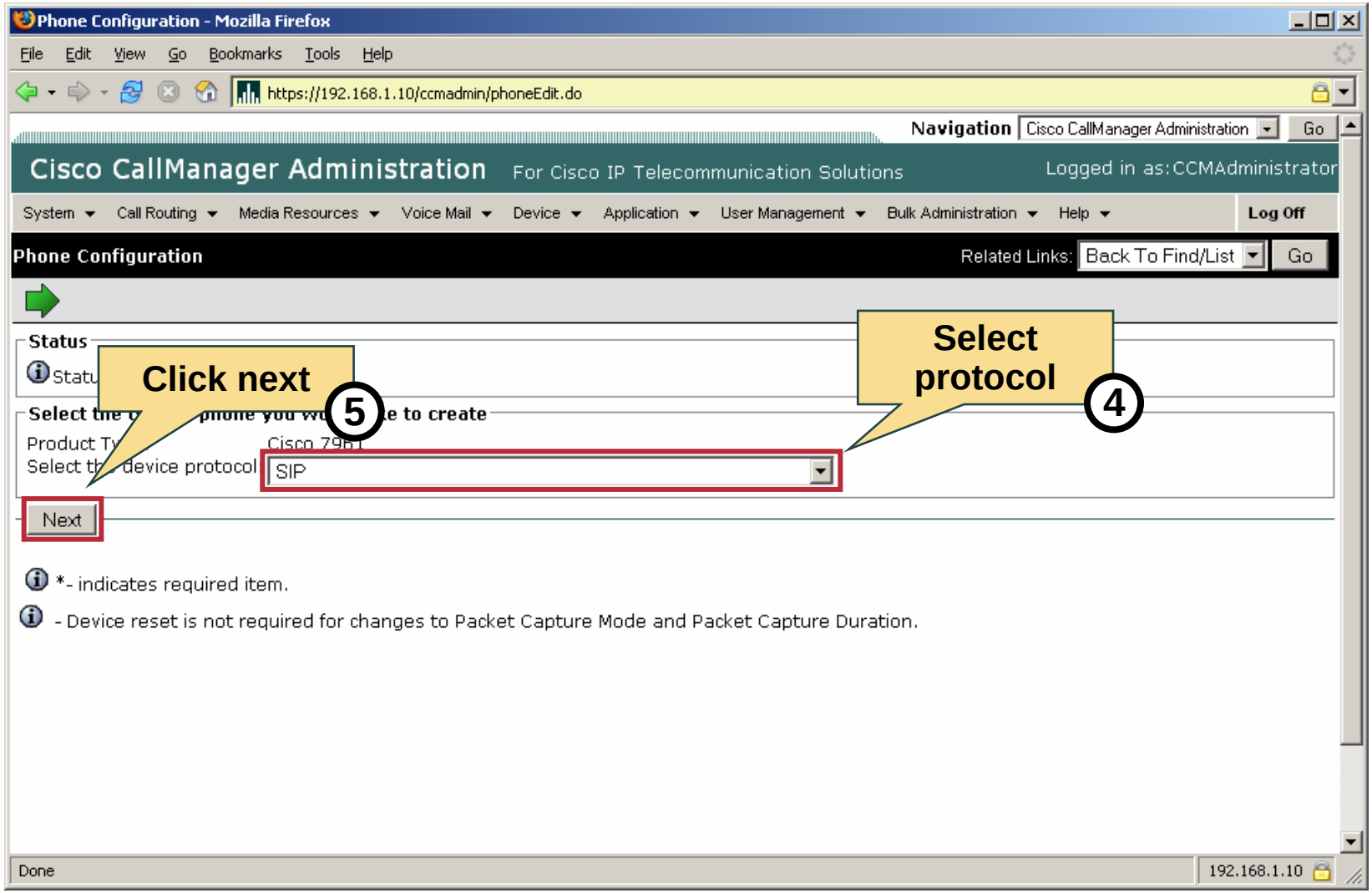
Configuring Cisco SIP Phones



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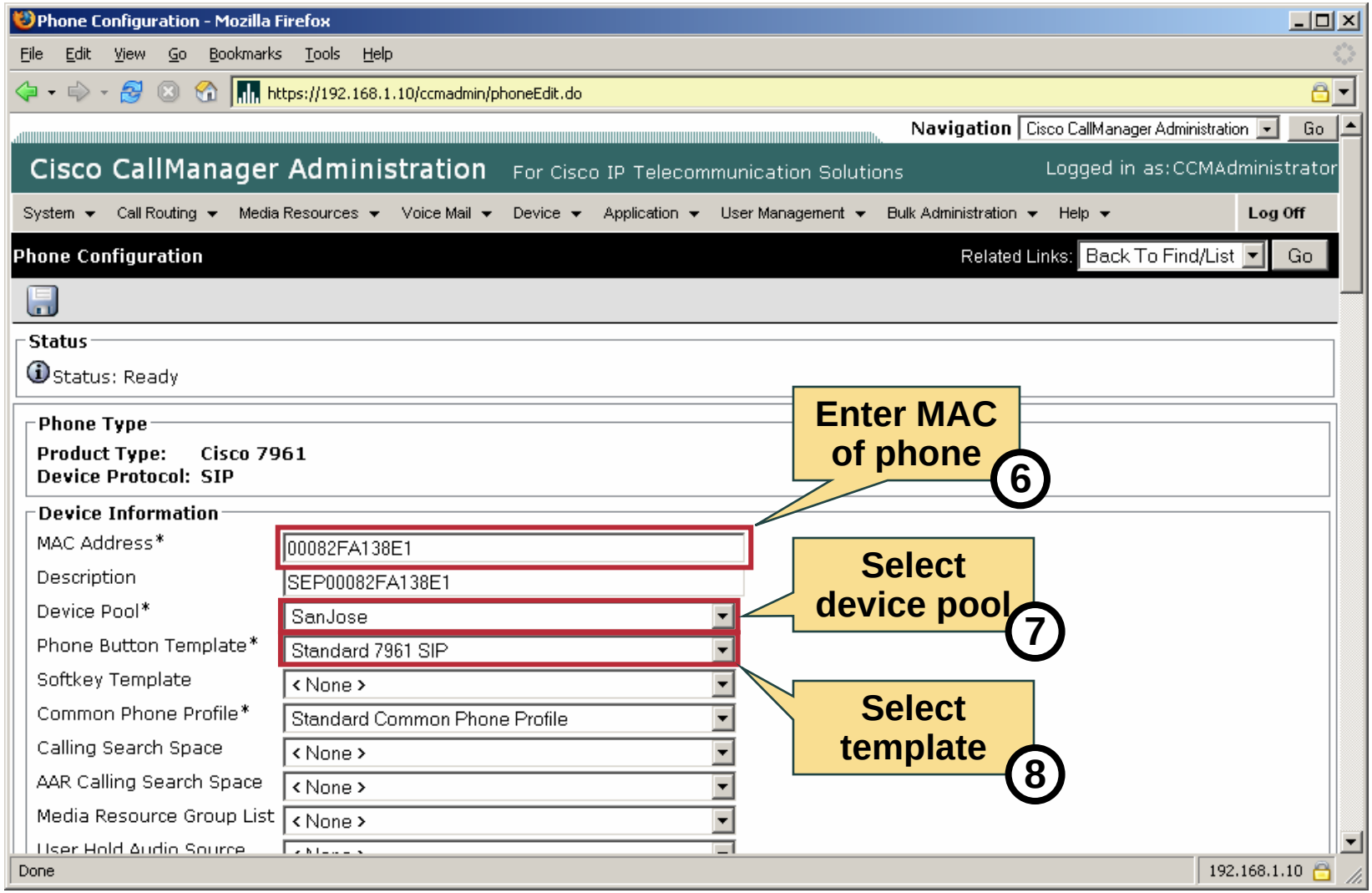
Configuring Cisco SIP Phones



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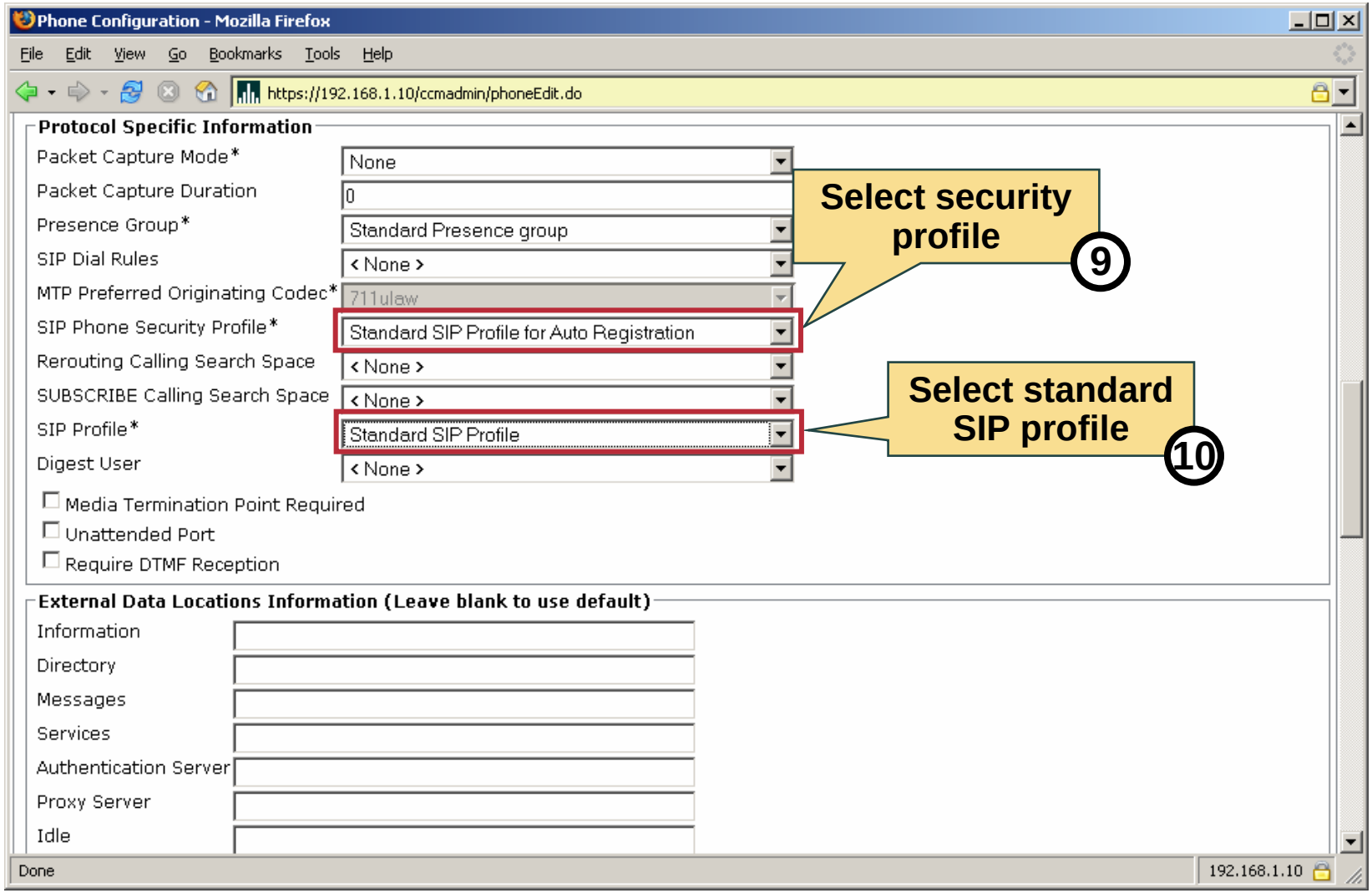
Configuring Cisco SIP Phones



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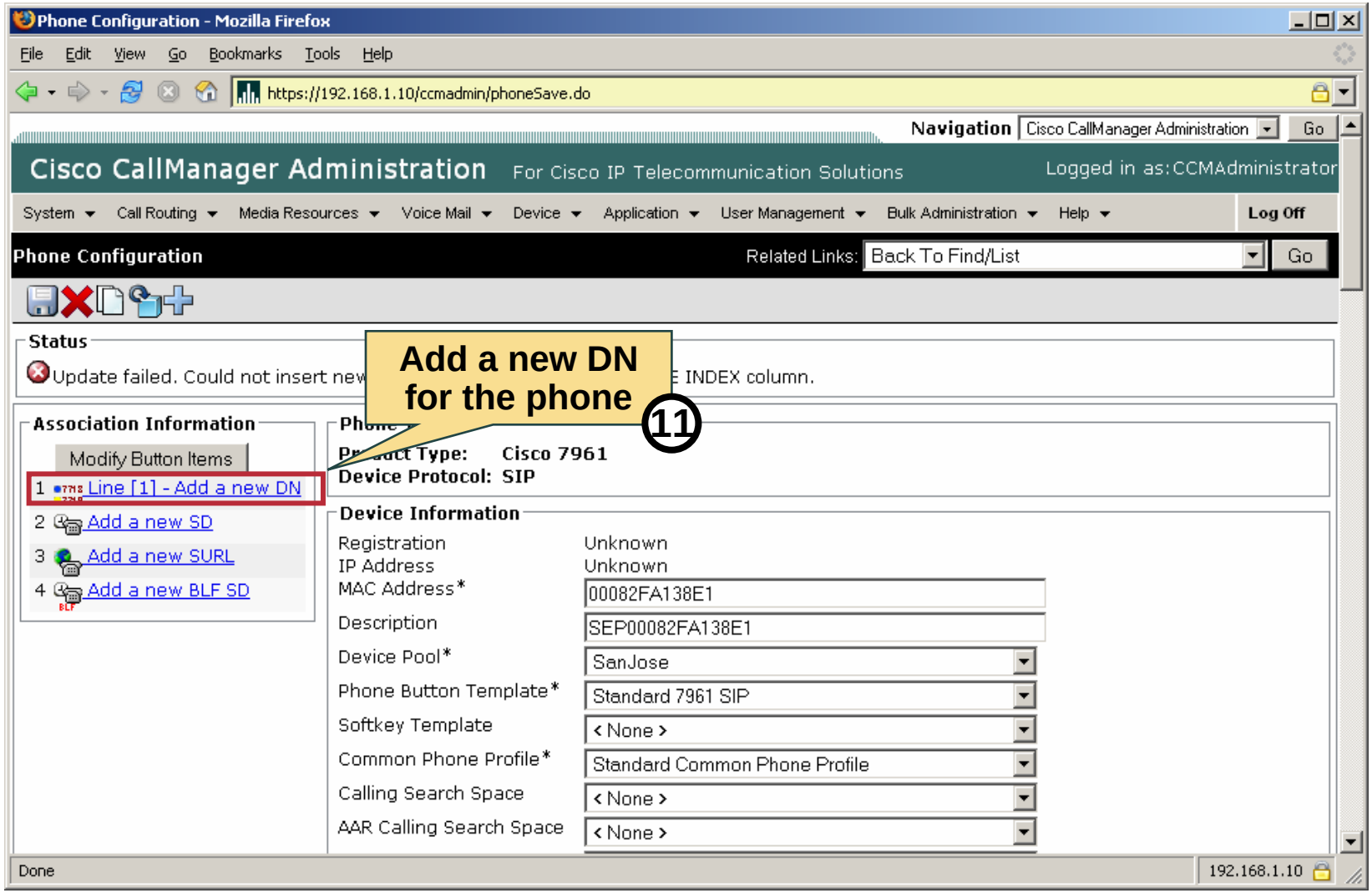
Configuring Cisco SIP Phones



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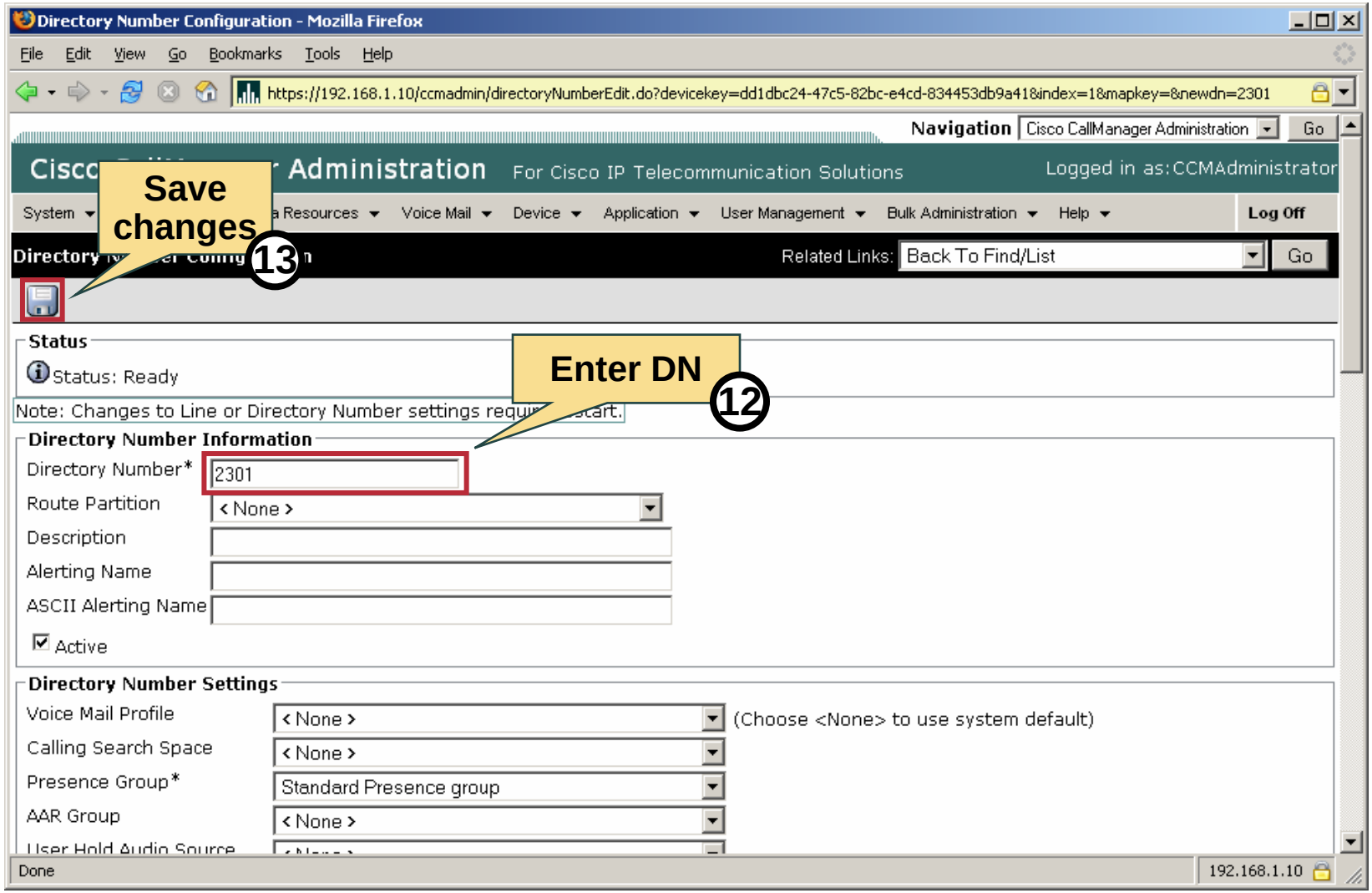
T. E.2

Configuring Cisco SIP Phones



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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.

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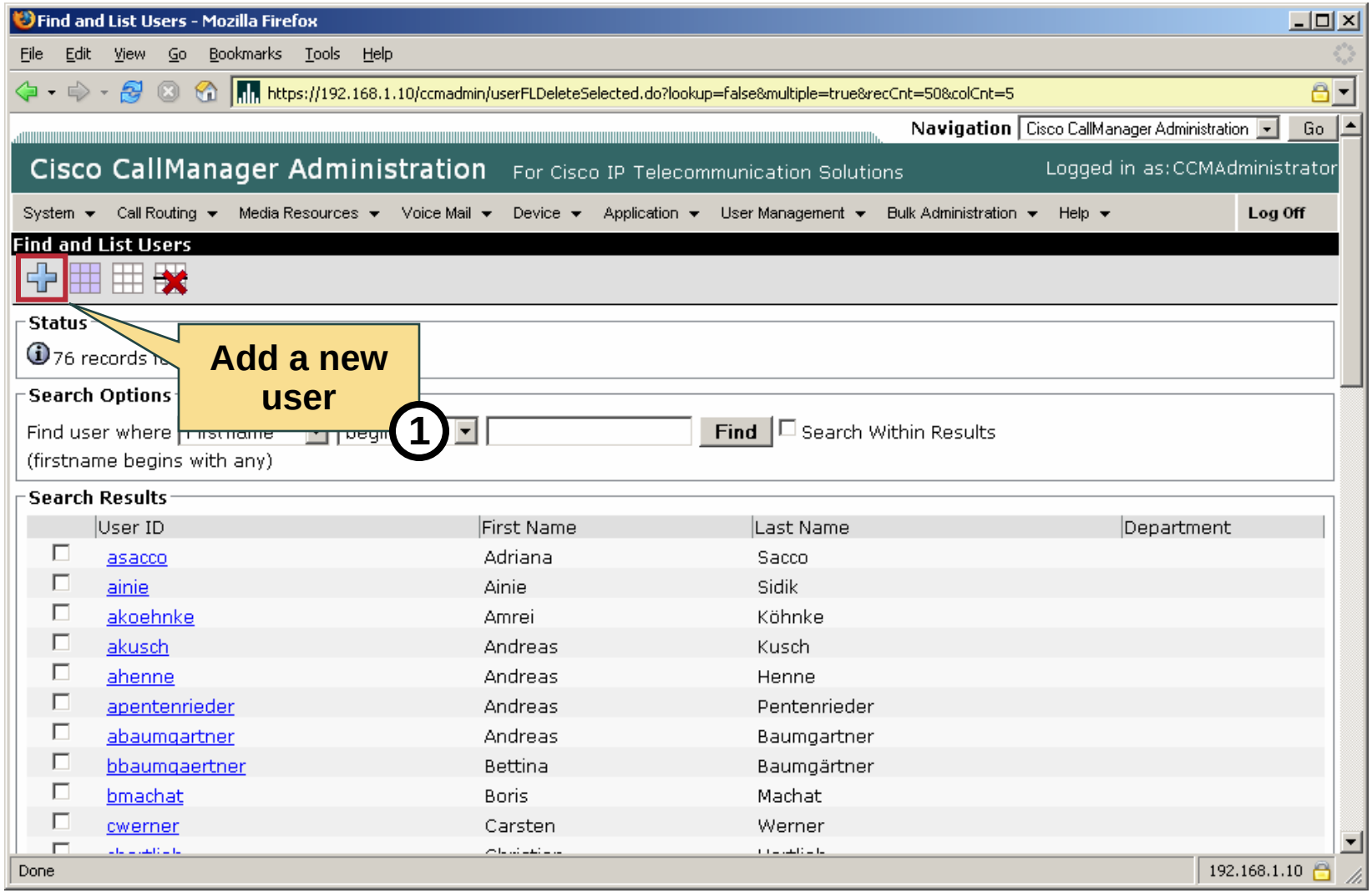
Configuring 3rd Party SIP Phones



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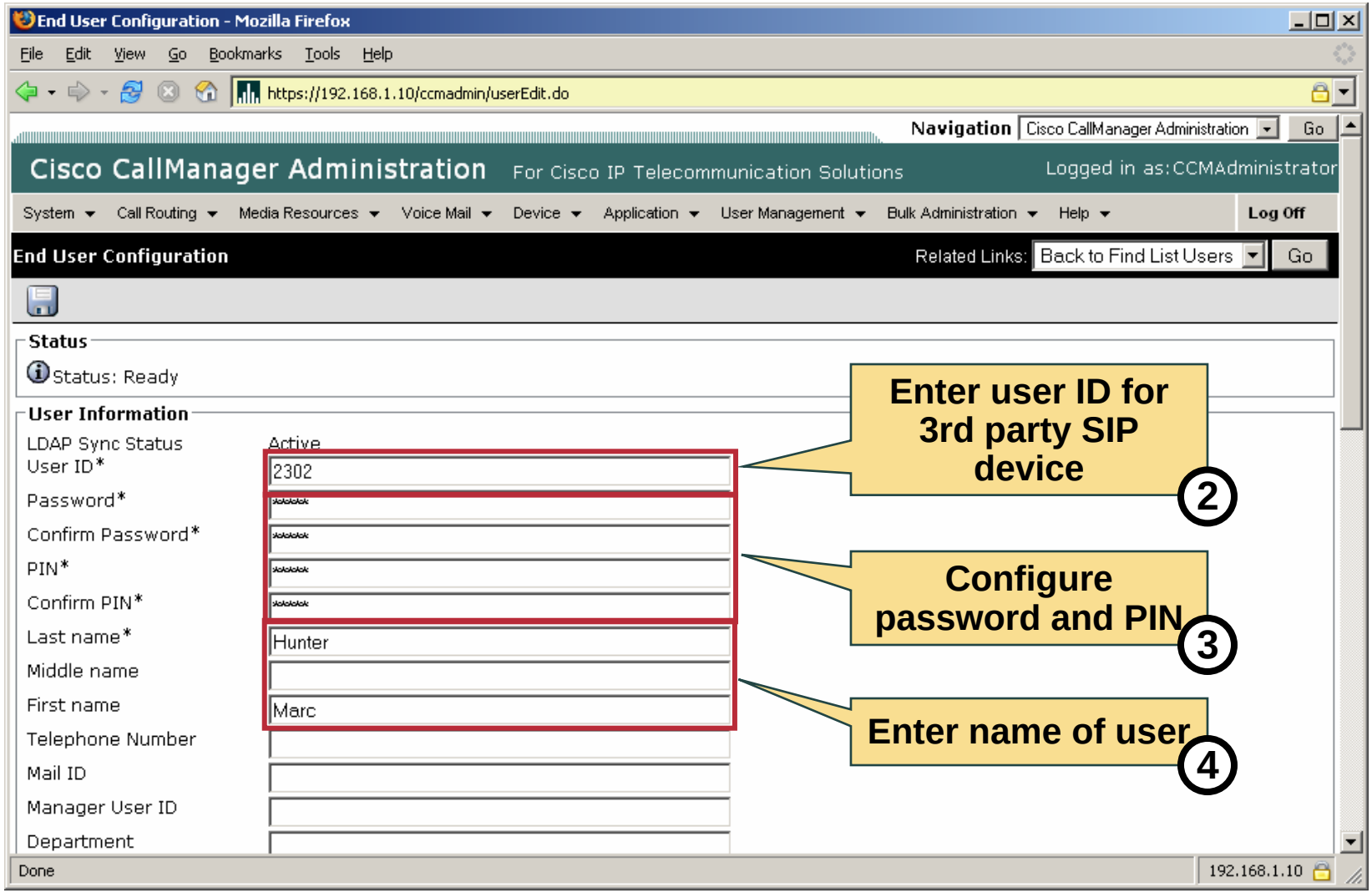
Configuring 3rd Party SIP Phones



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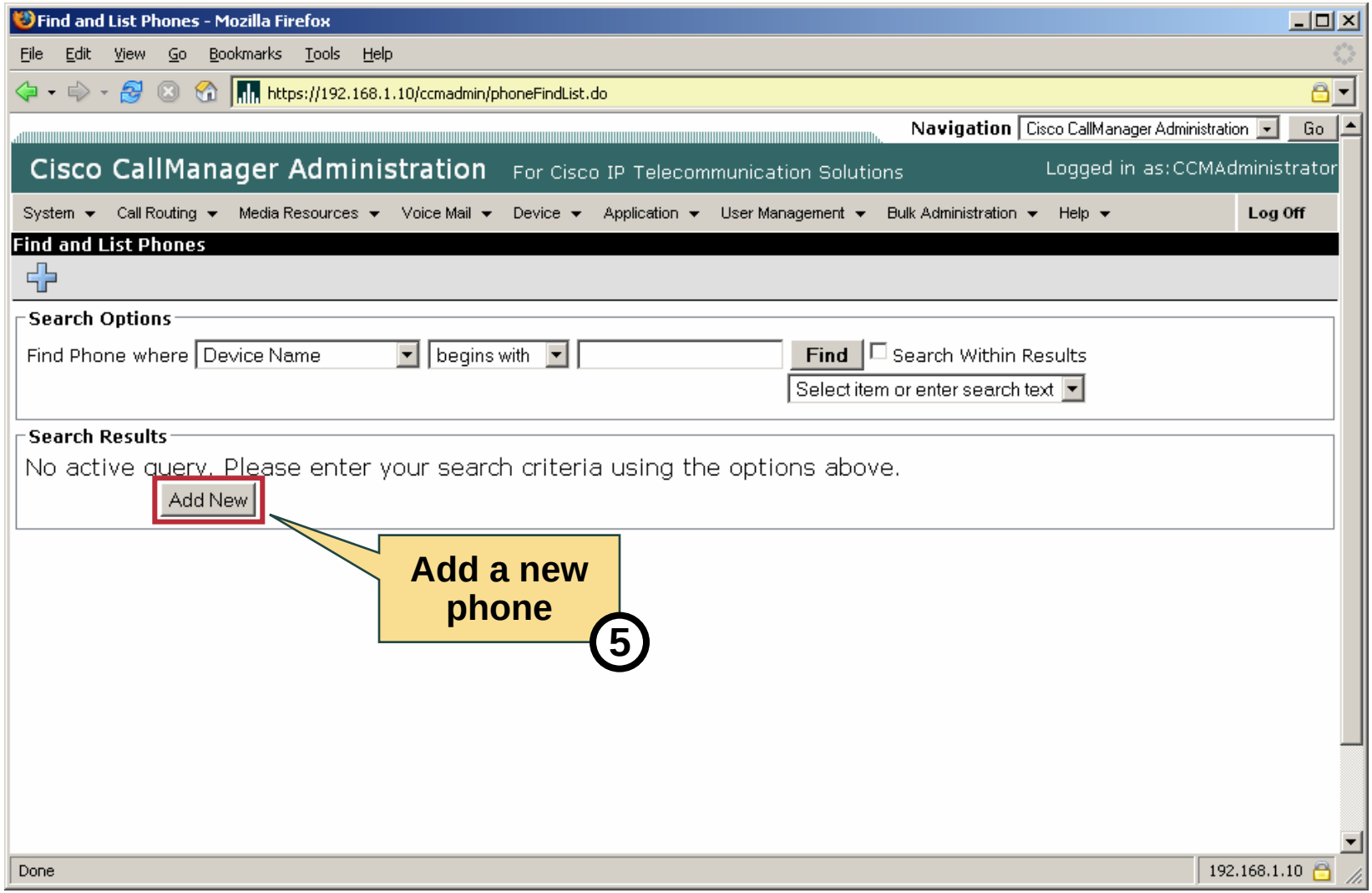
Configuring 3rd Party SIP Phones



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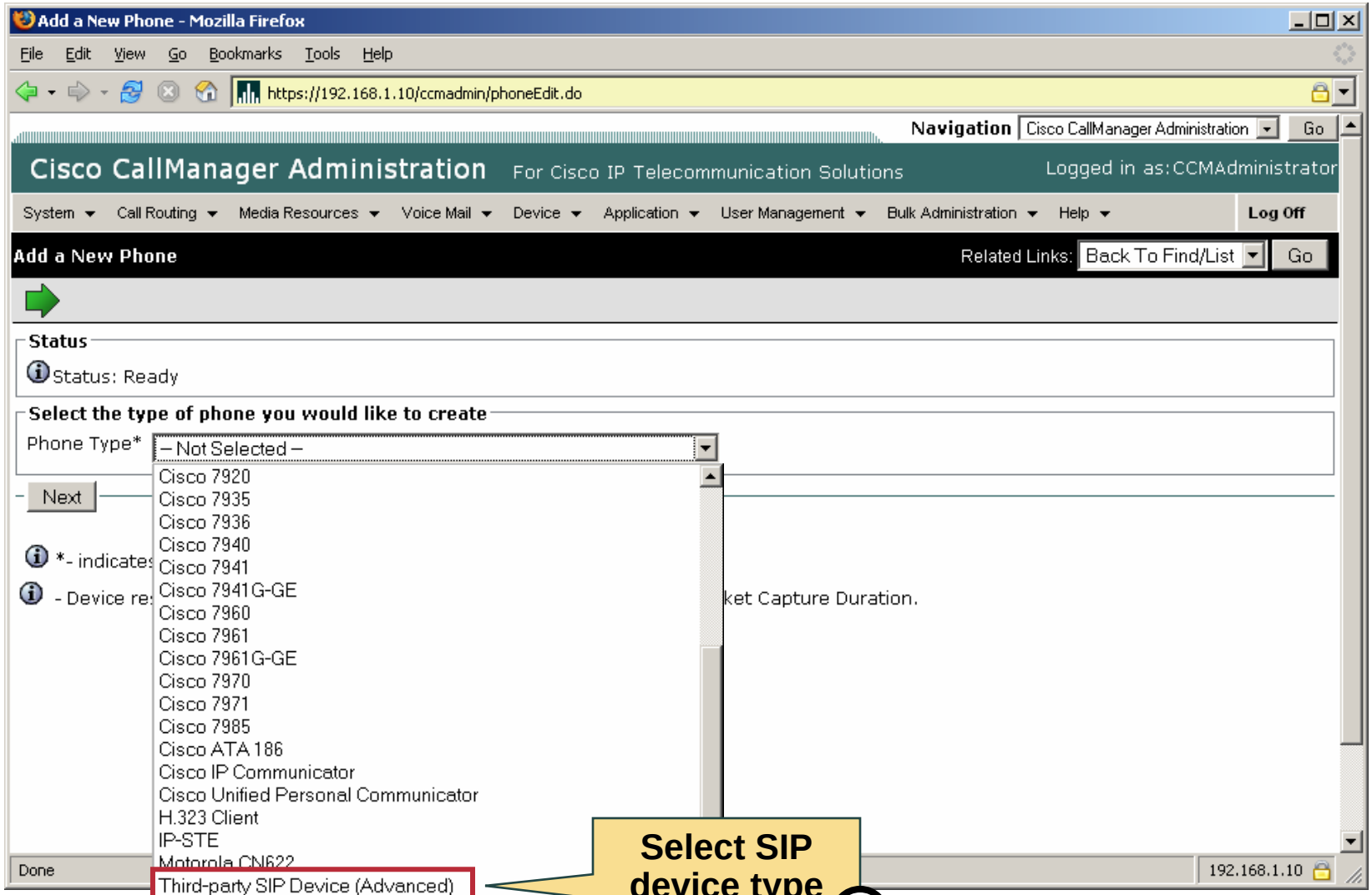
Configuring 3rd Party SIP Phones



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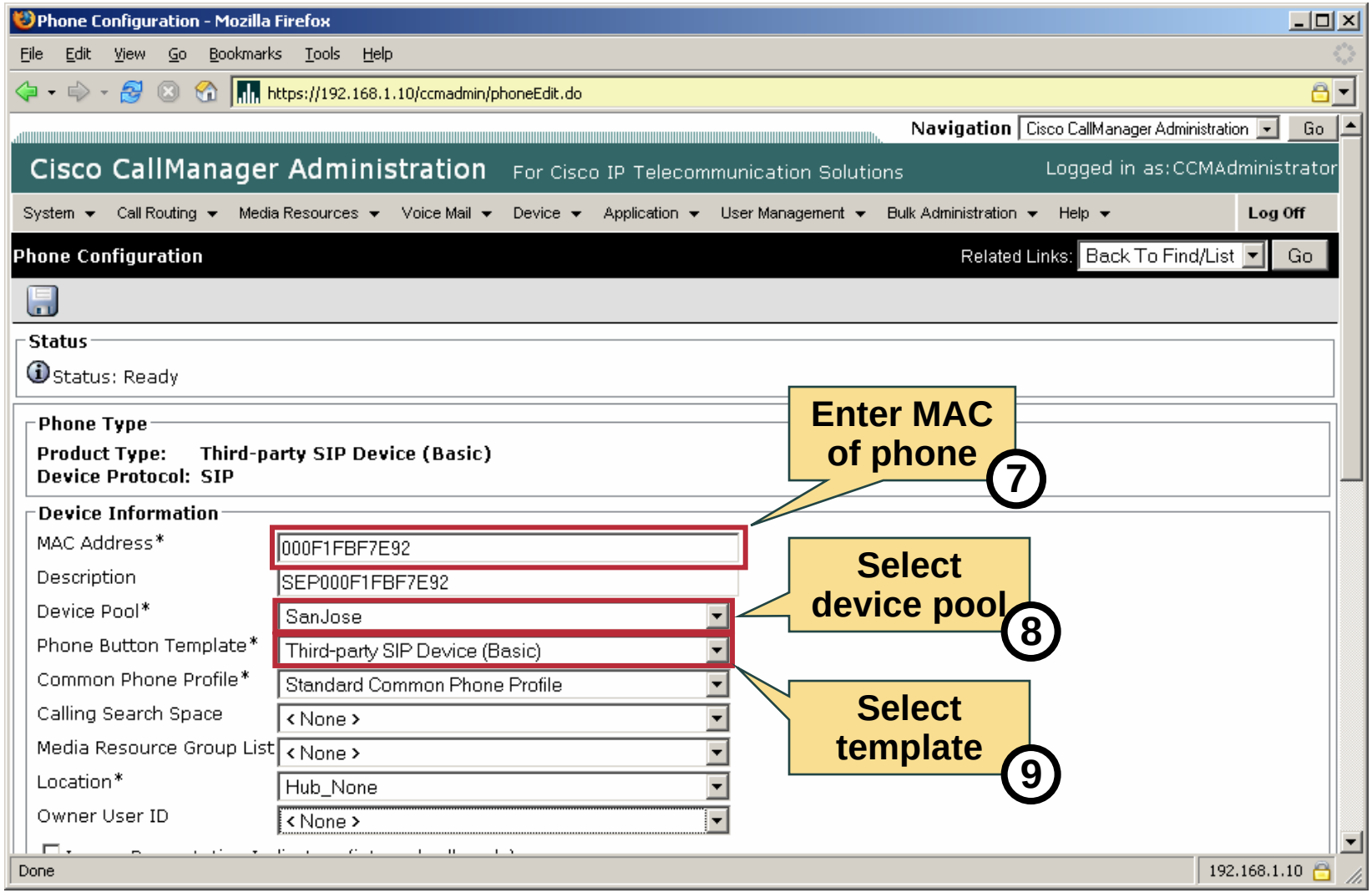
Configuring 3rd Party SIP Phones



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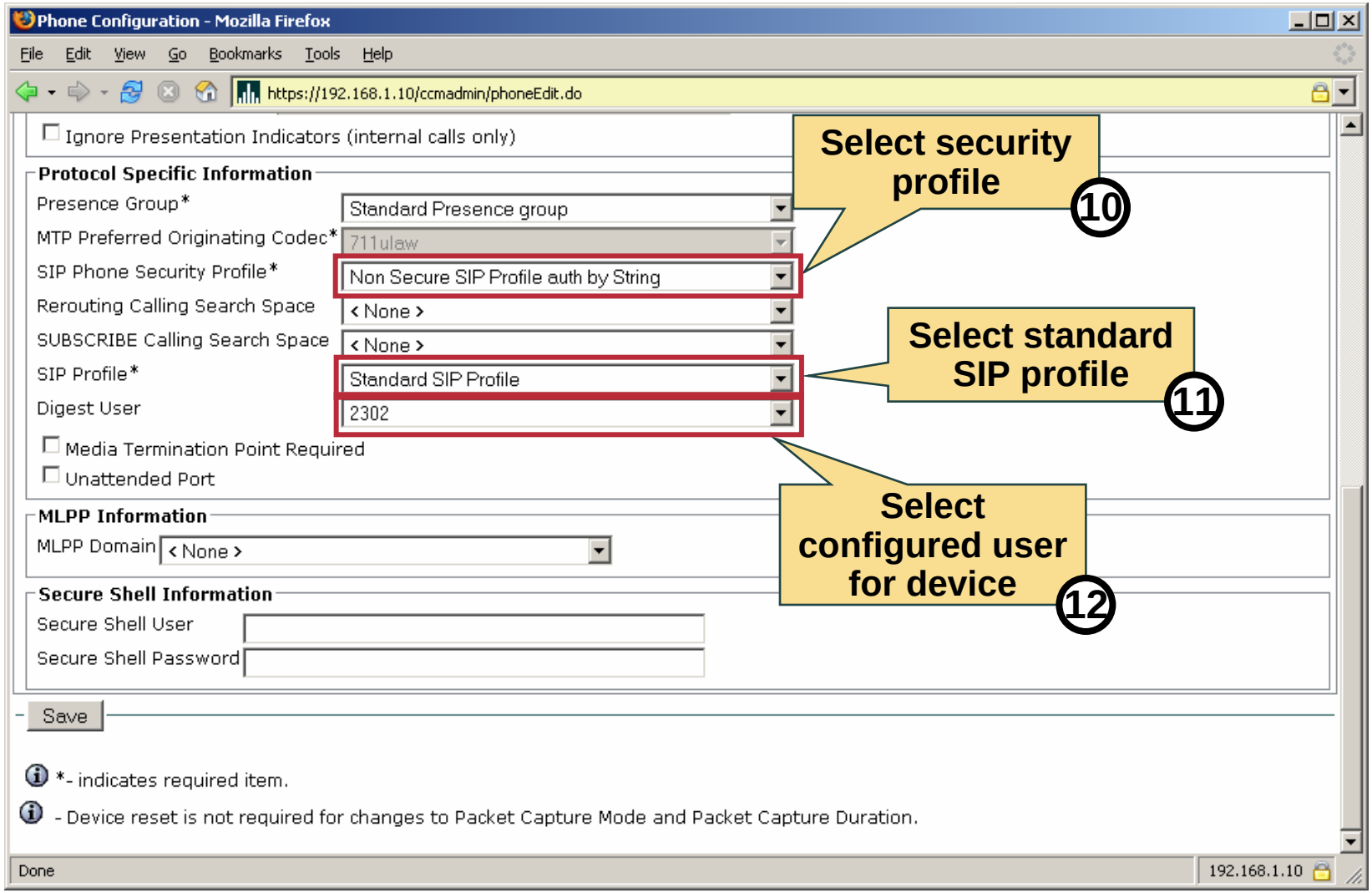
Configuring 3rd Party SIP Phones



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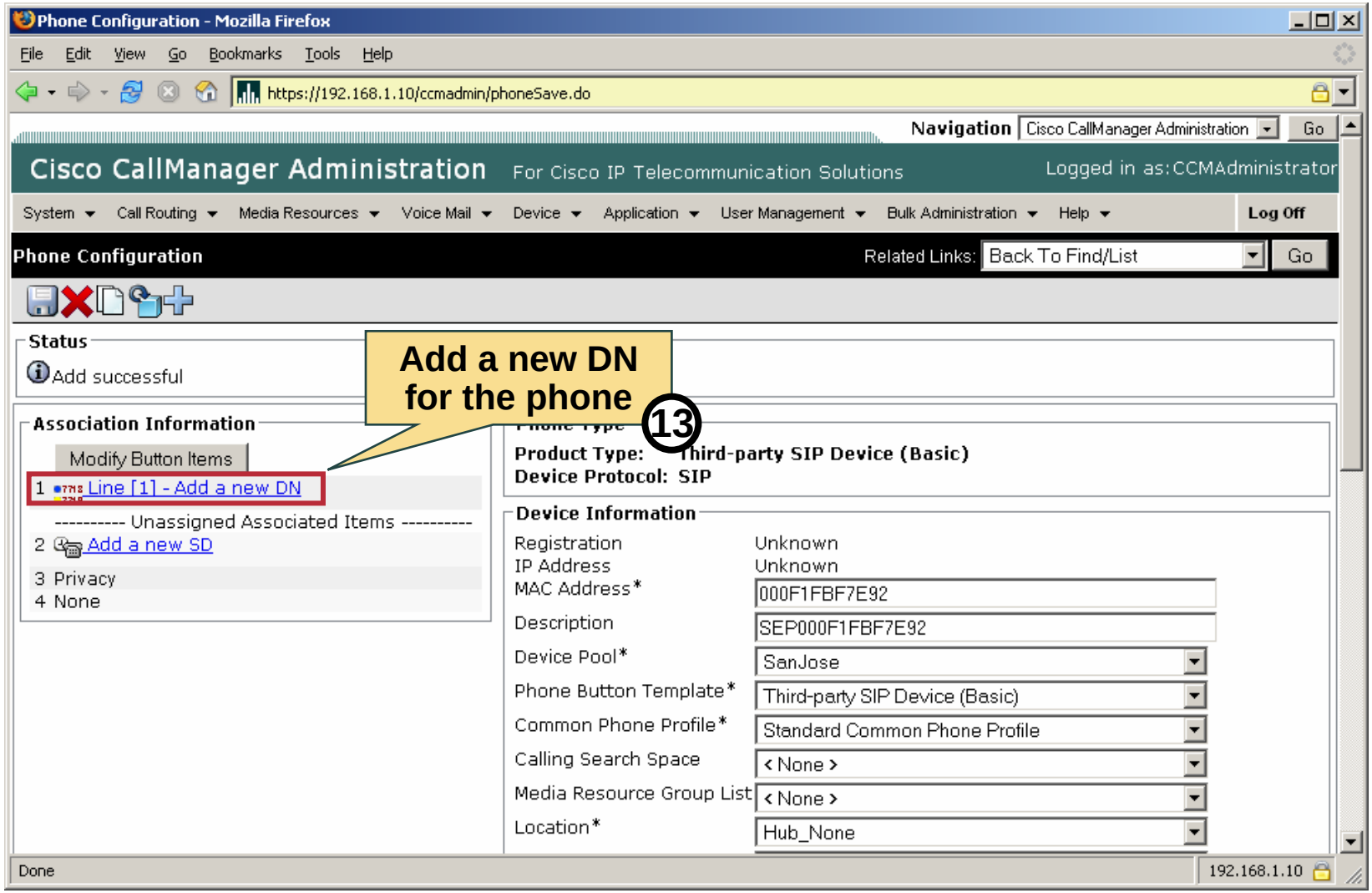
Configuring 3rd Party SIP Phones



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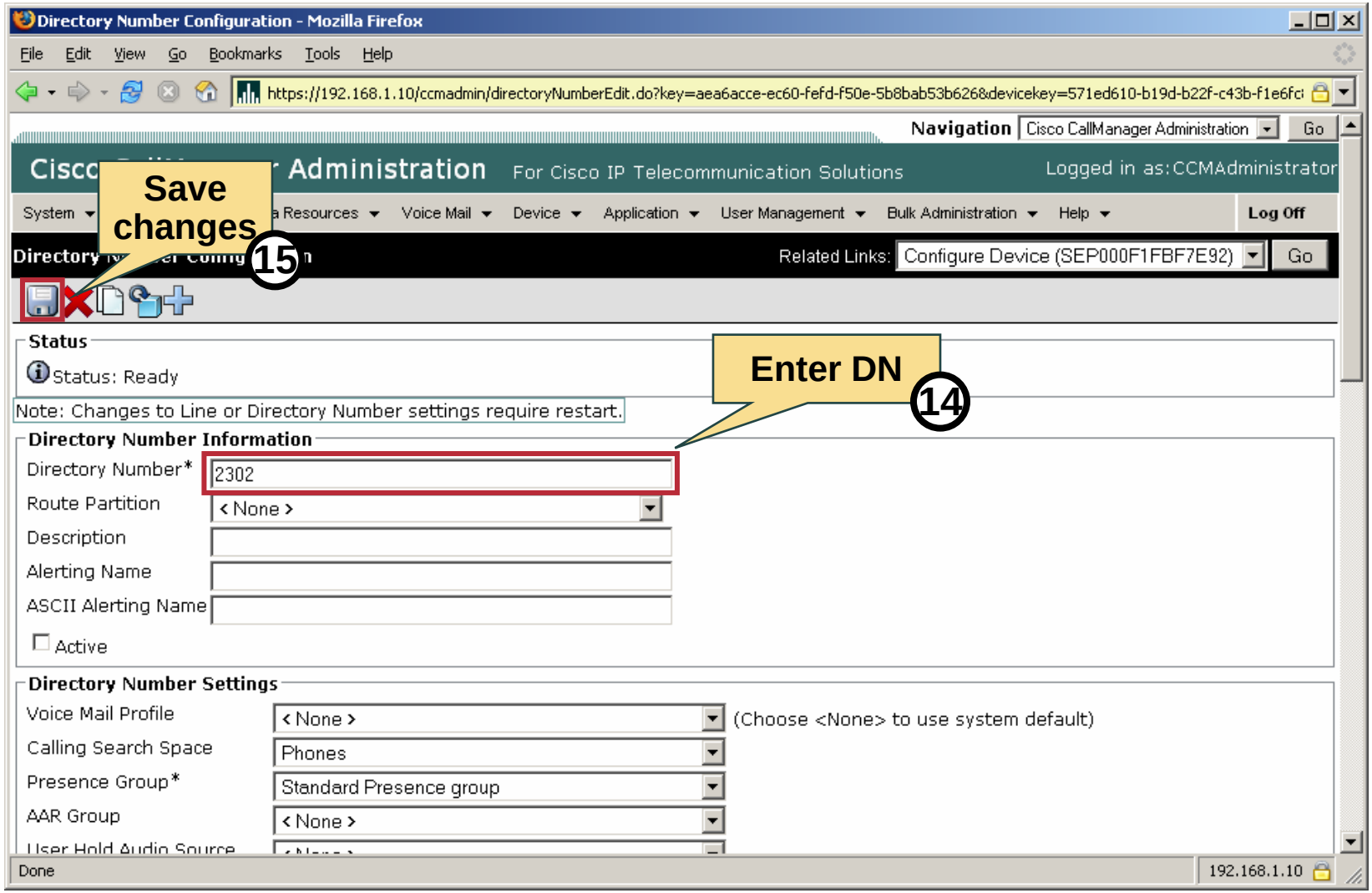
Configuring 3rd Party SIP Phones



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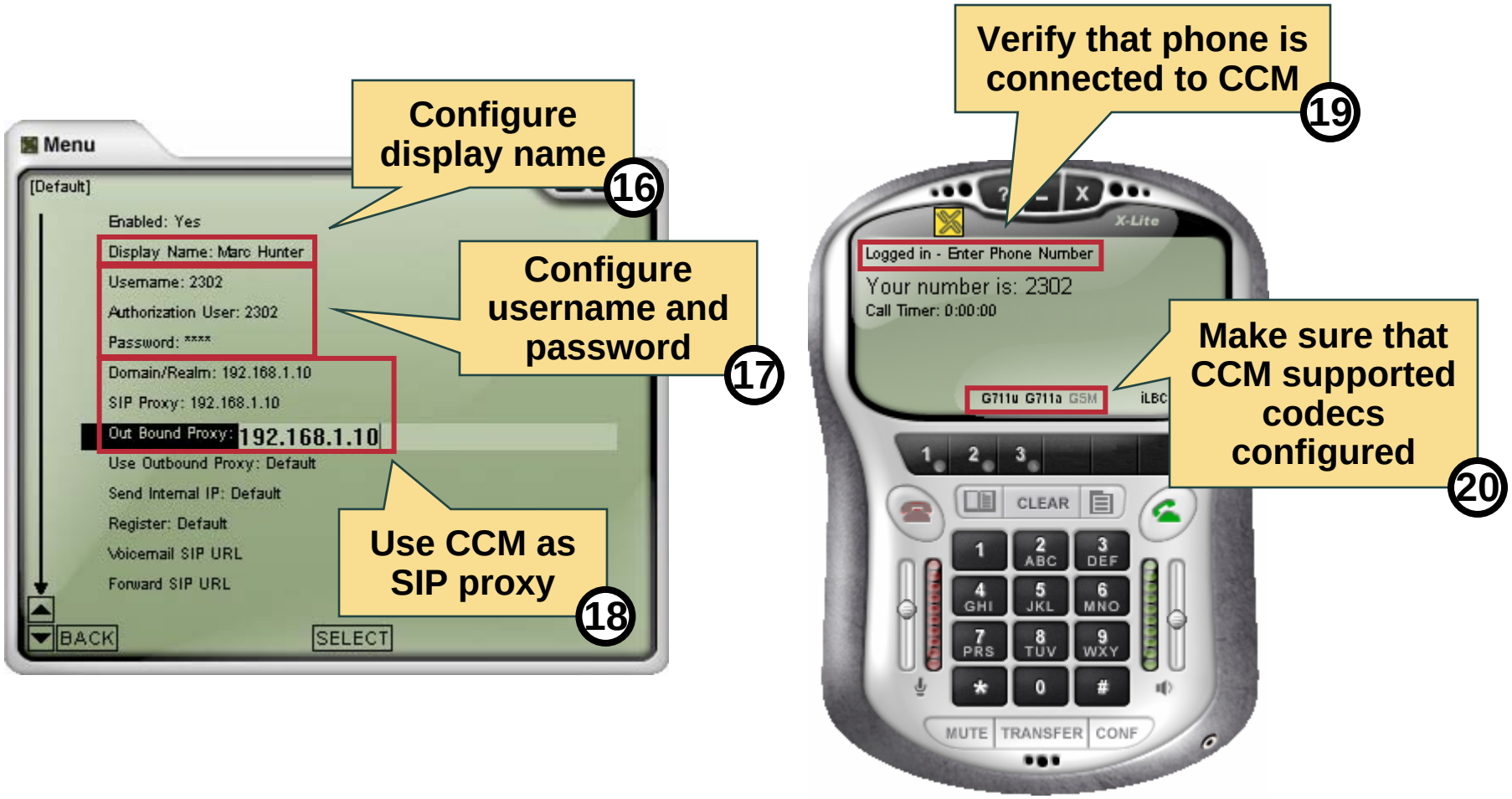
Configuring 3rd Party SIP Phones



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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.

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Deploying Cisco Unified CallManager 5.0 Endpoints

Managing Endpoints

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Objectives

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

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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.**

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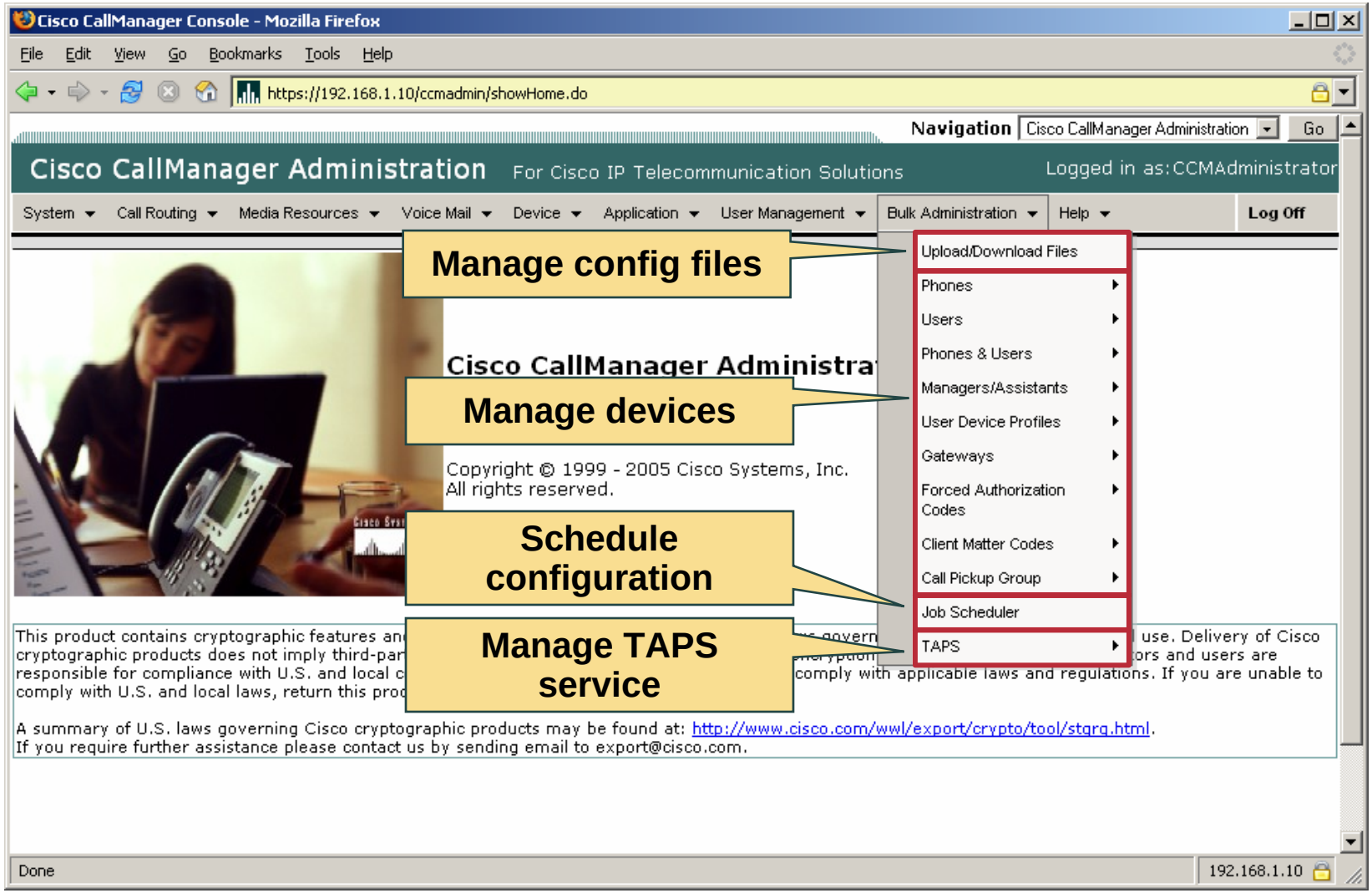
Bulk Administration Tool (BAT) Overview



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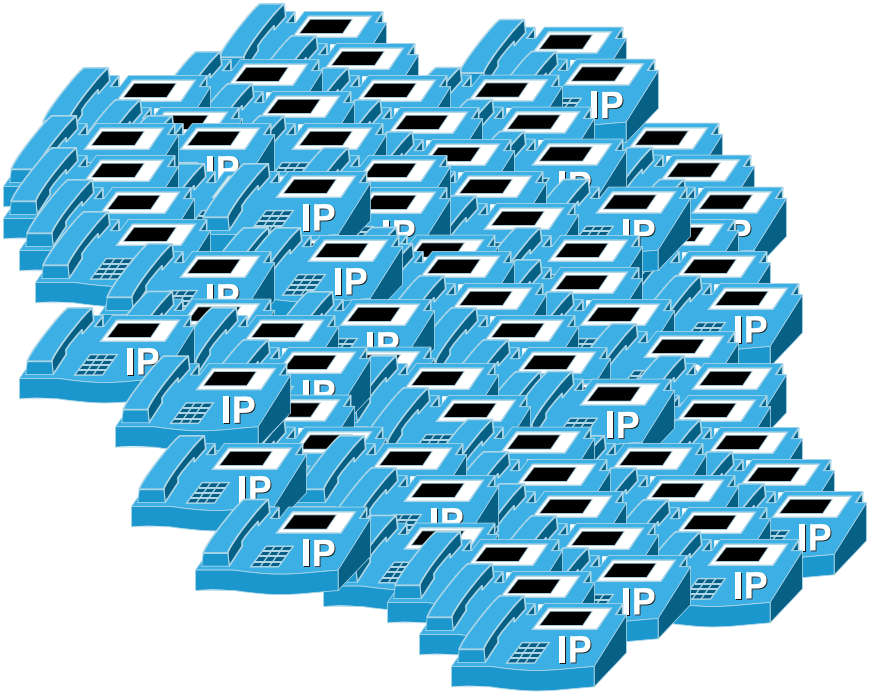
Using BAT

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



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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.

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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**

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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.

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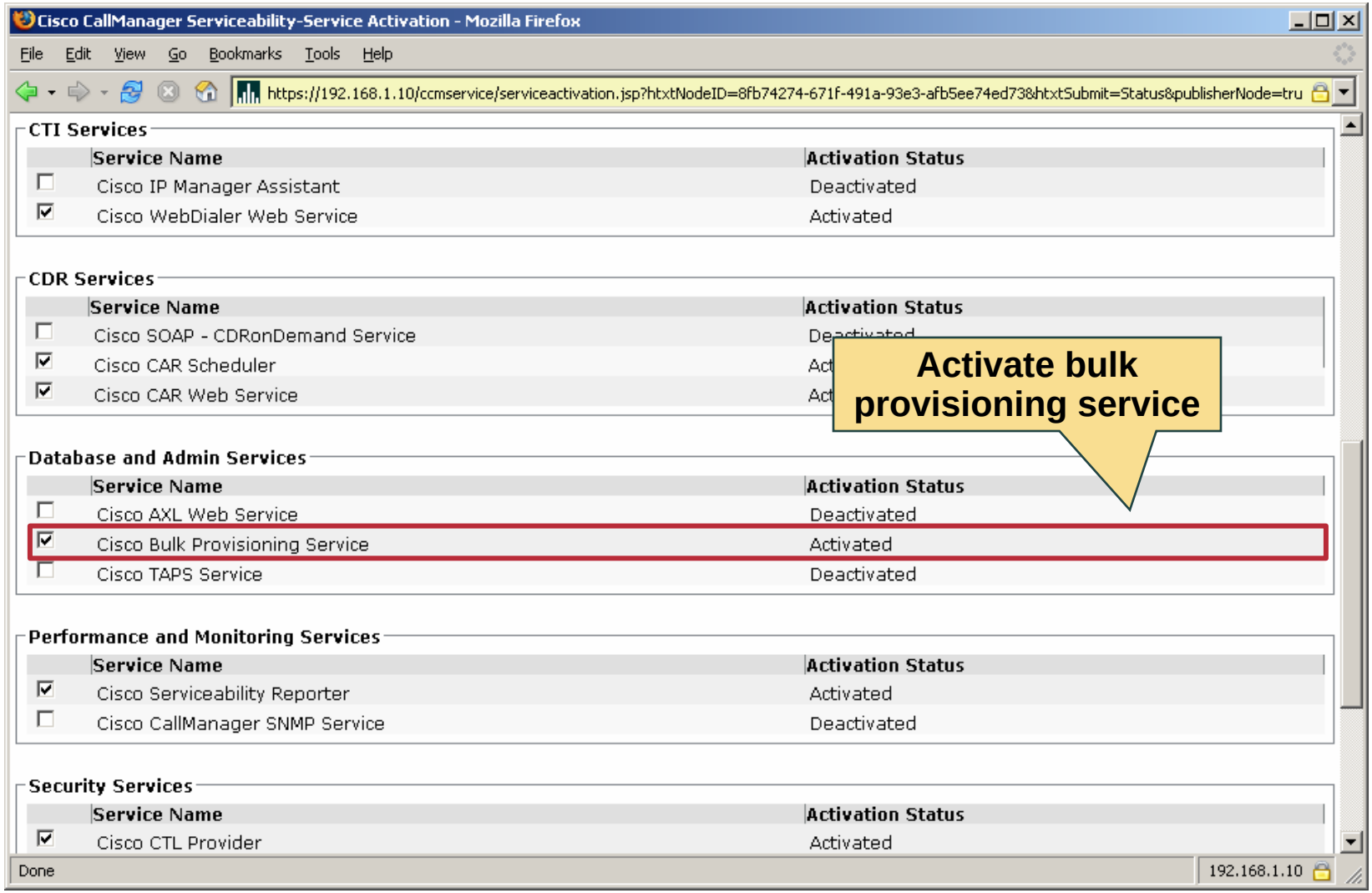
Using BAT



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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.**

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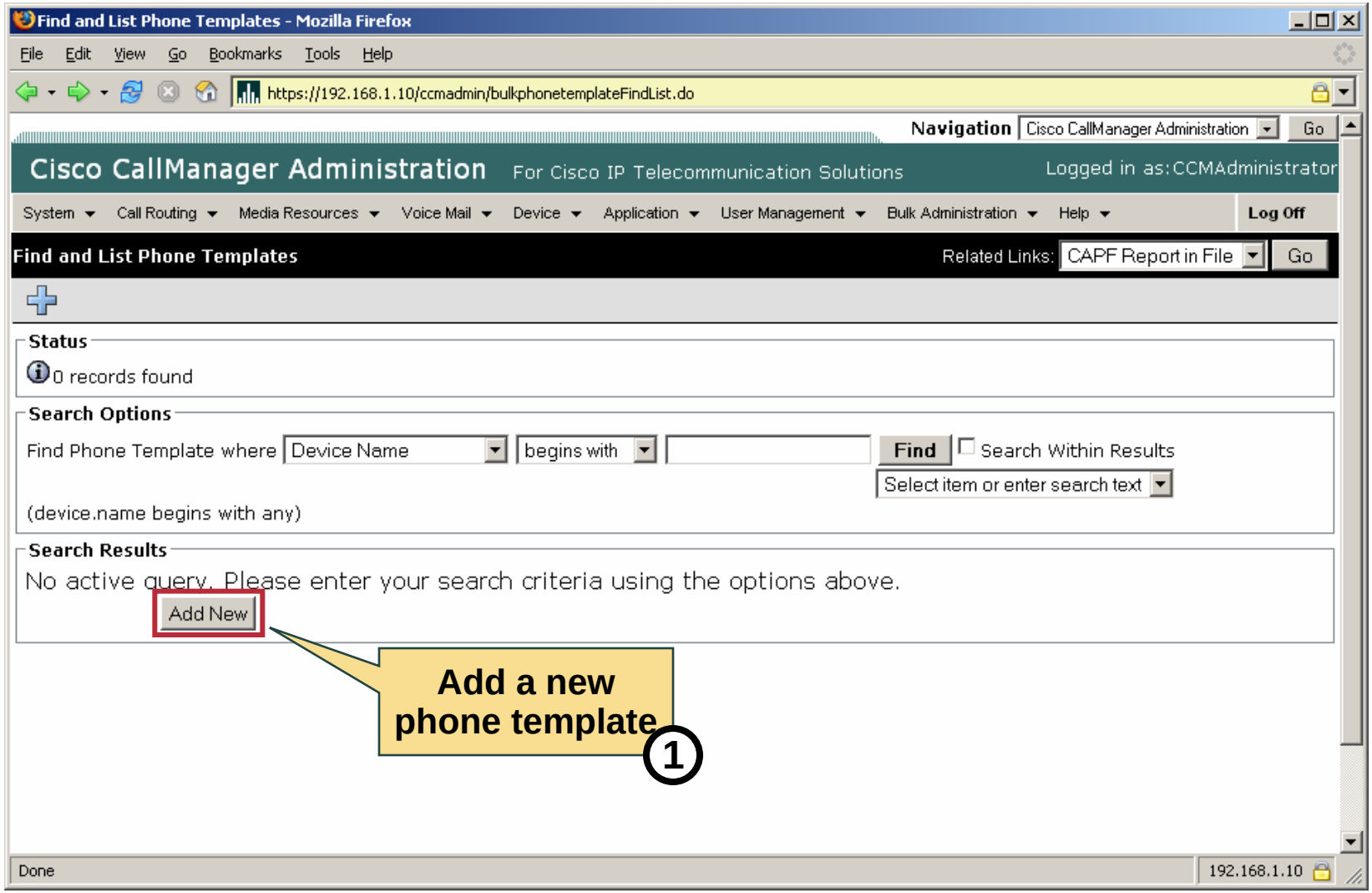
Configuring BAT Phone Templates



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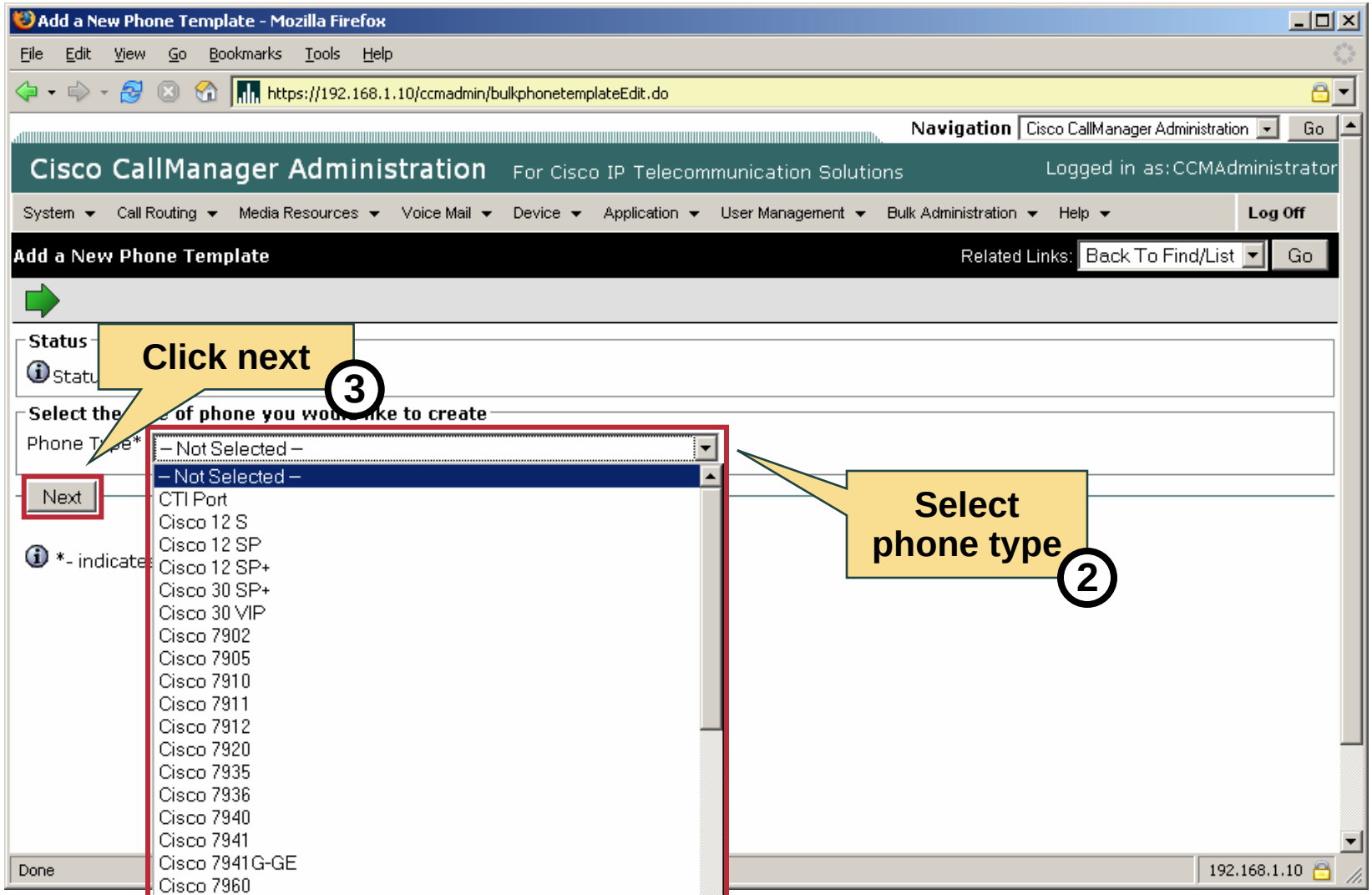
Configuring BAT Phone Templates



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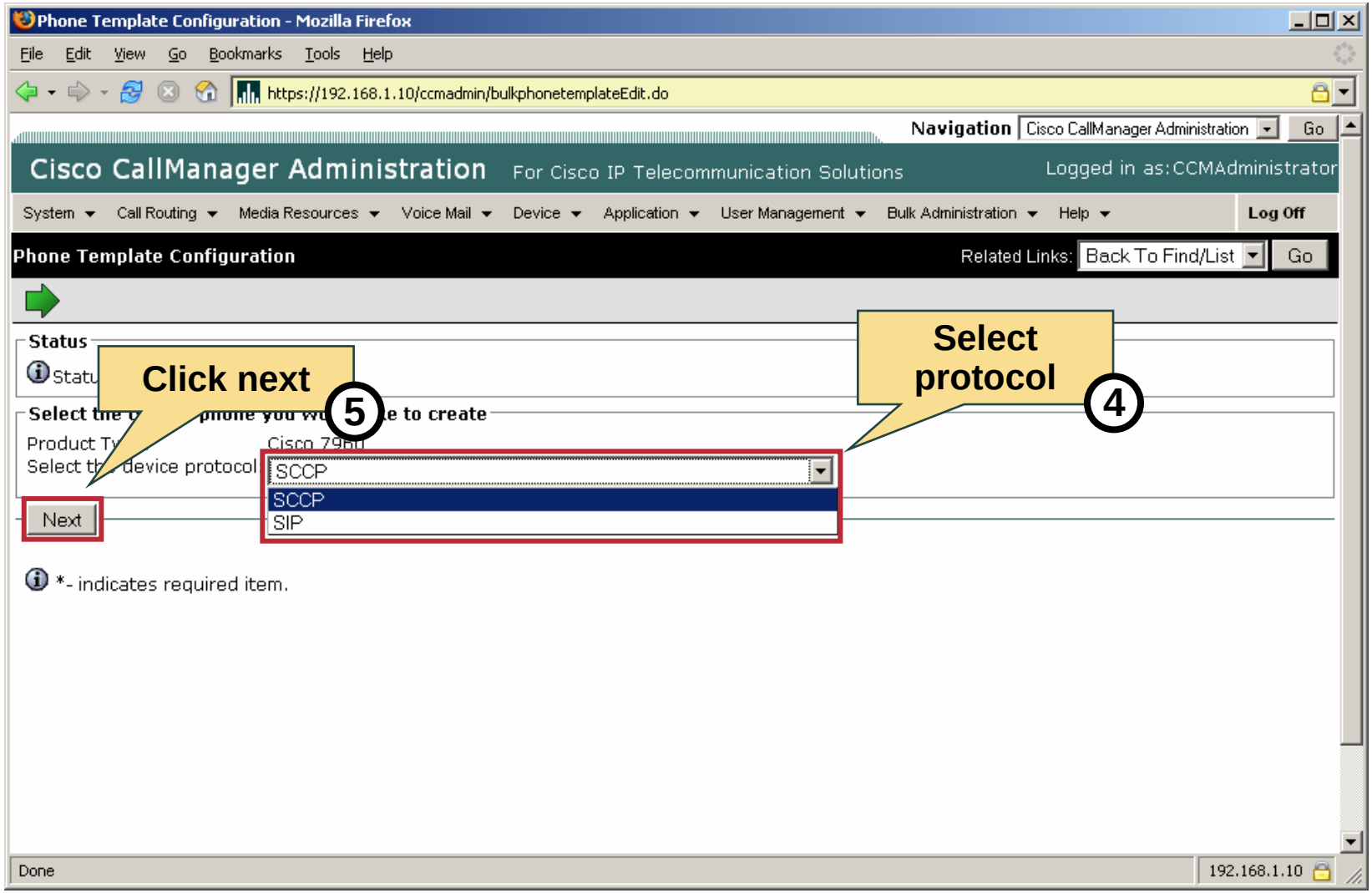
Configuring BAT Phone Templates



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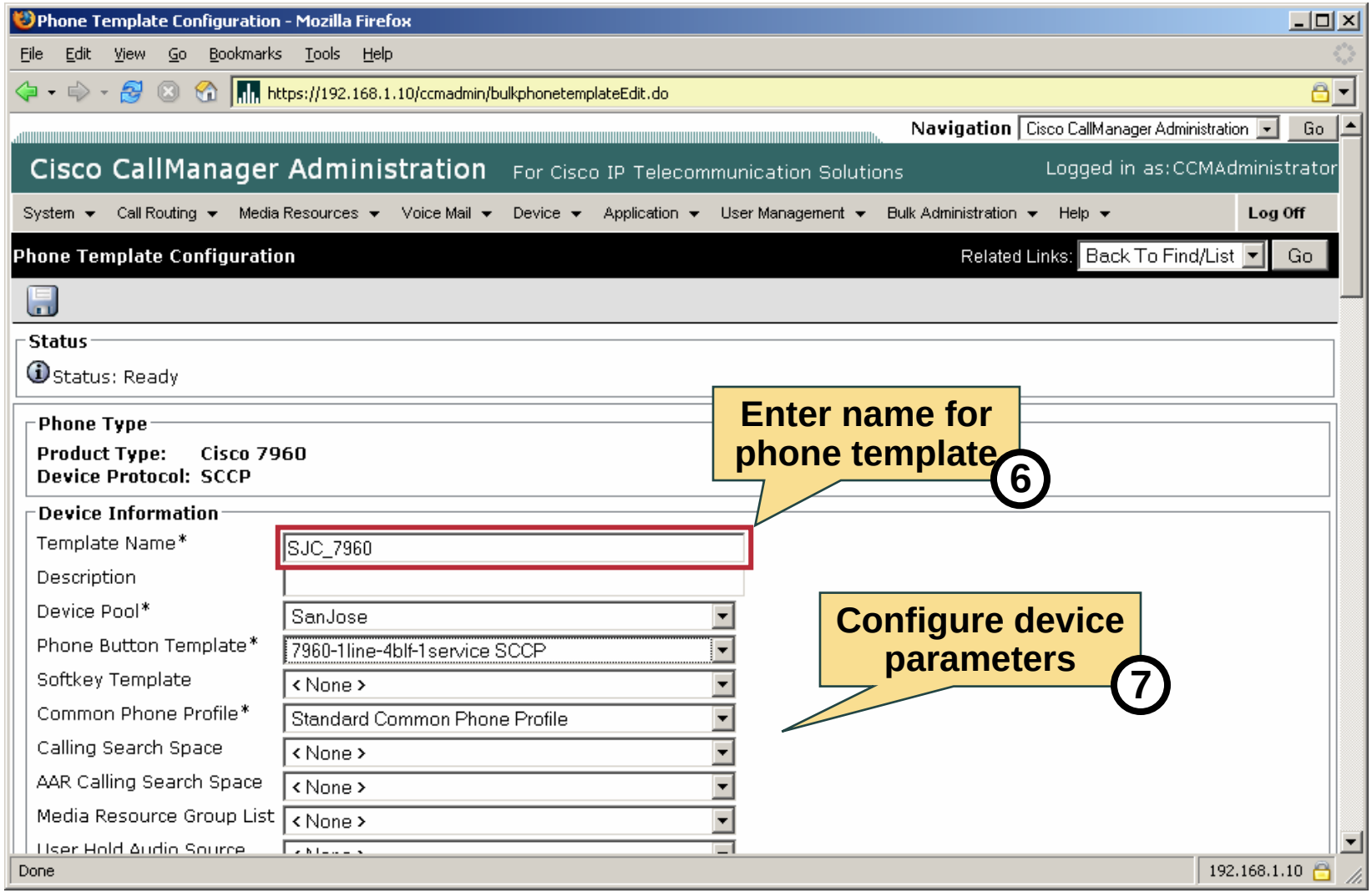
Configuring BAT Phone Templates



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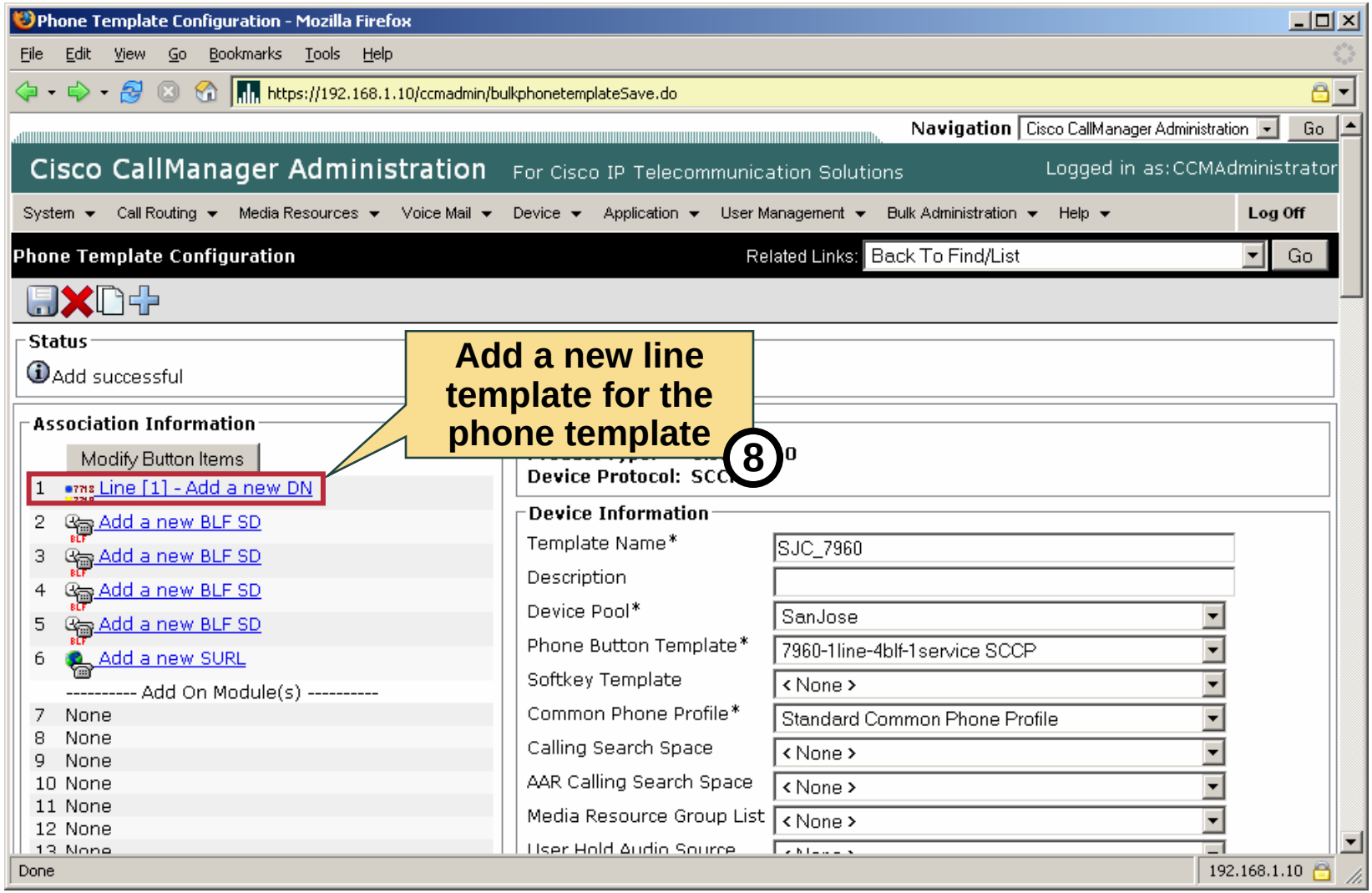
Configuring BAT Phone Templates



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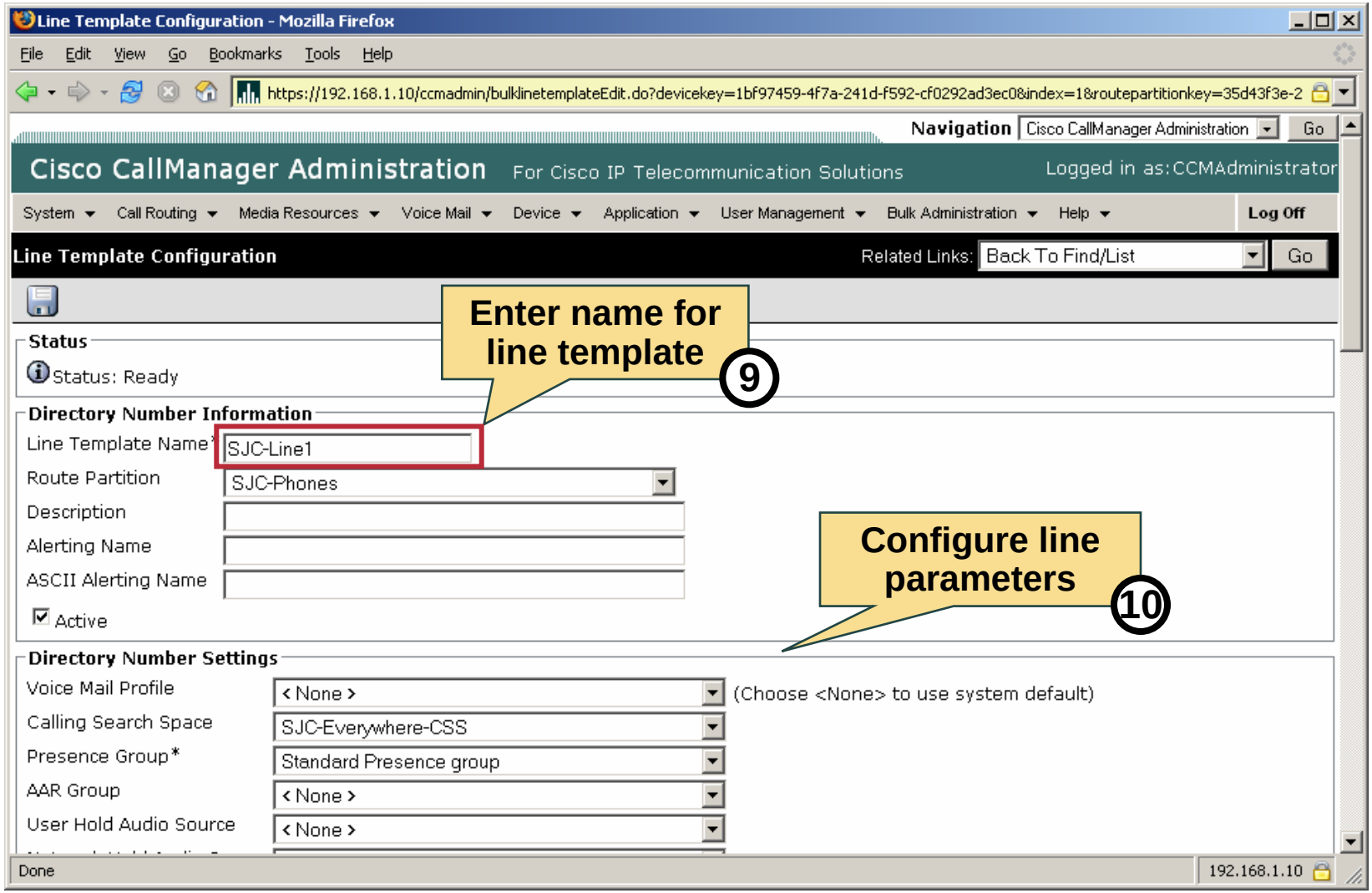
Configuring BAT Phone Templates



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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.**

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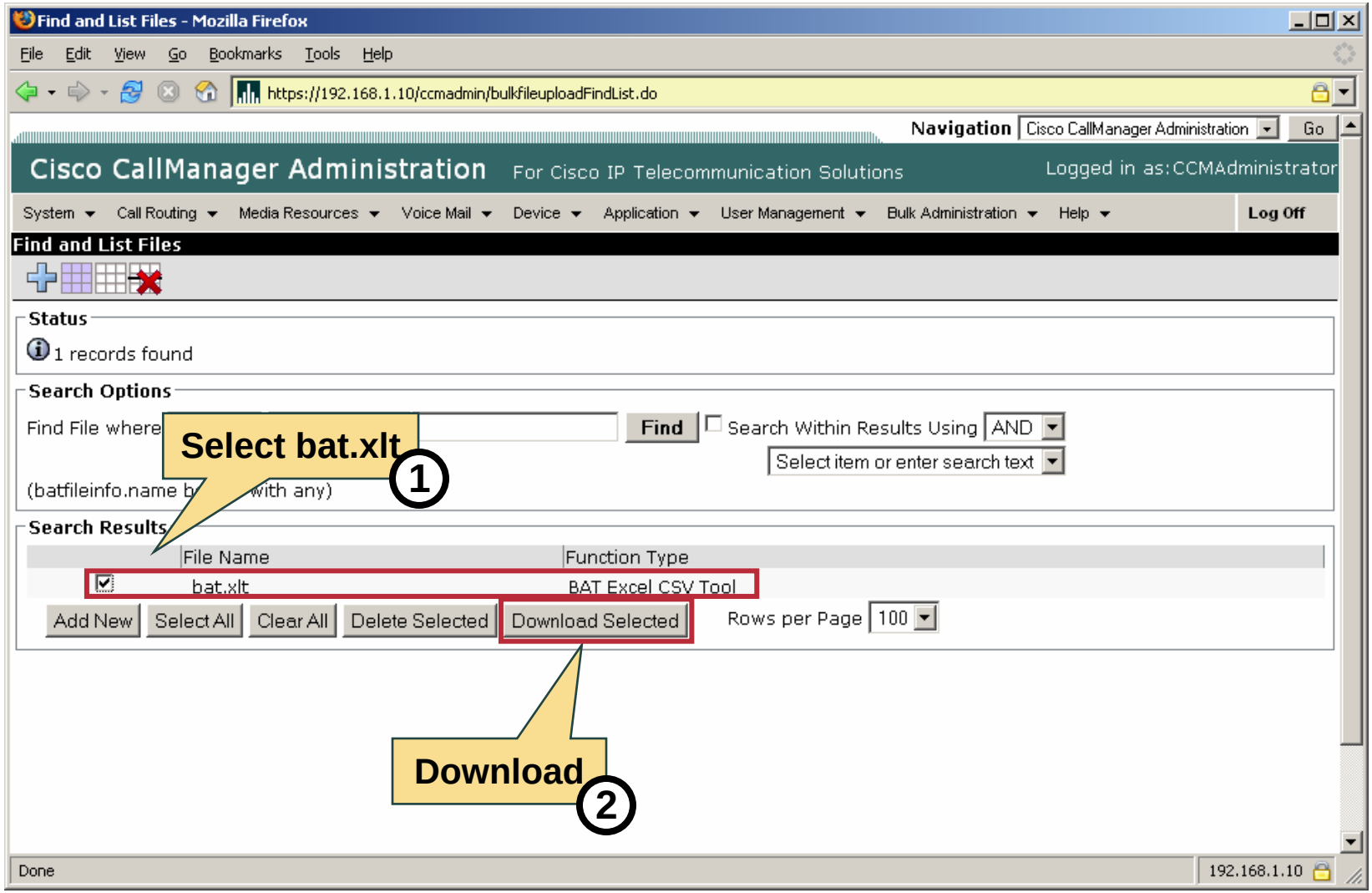
Creating CSV Files



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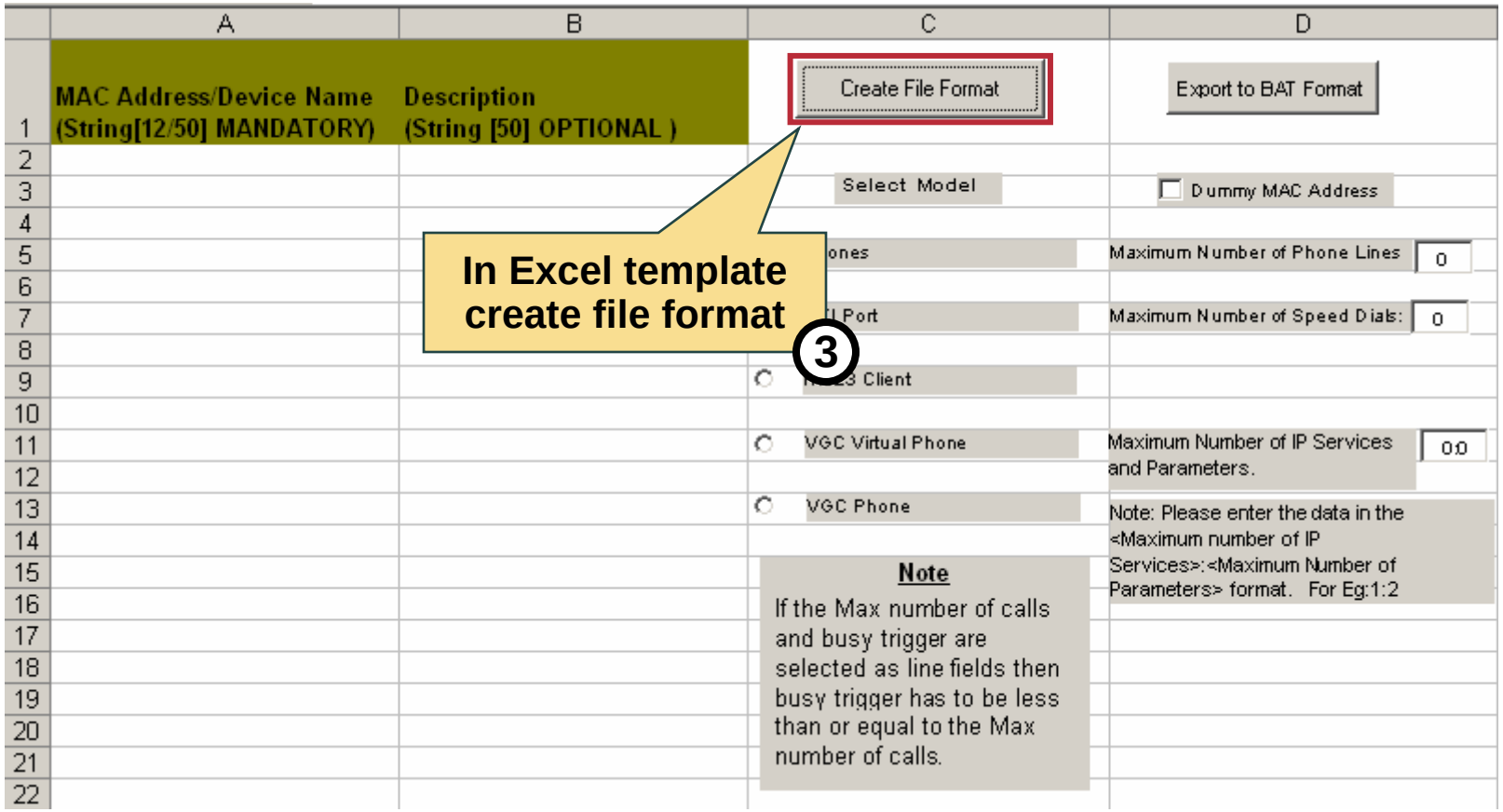
Creating CSV Files



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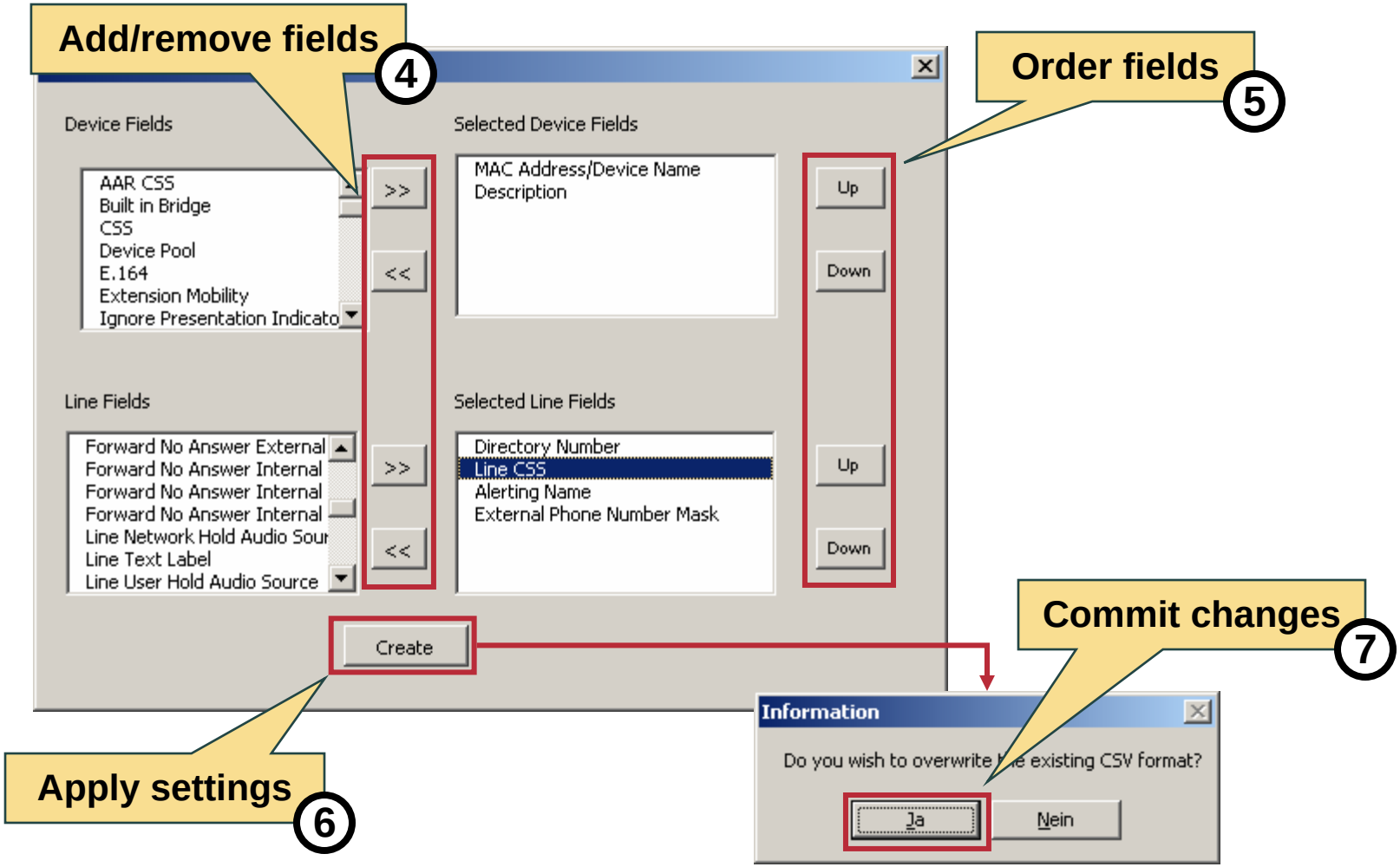
Creating CSV Files



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Creating CSV Files

B	C	D	E	F
Description (String [50] OPTIONAL)	**Directory Number 1 (Integer [50] MANDATORY)**	**Line CSS 1 (String [50] OPTIONAL)**	**Alerting Name 1 (String [50] OPTIONAL)**	**External Phone Number Mask 1 (String [50] OPTIONAL)**
SJC 2600	2600	SJC-Everywhere-CSS	Jimi Hendrix	1408555XXXX
SJC 2601	2601	SJC-Everywhere-CSS	Pete Townshend	1408555XXXX
SJC 2602	2602	SJC-Everywhere-CSS	Chuck Berry	1408555XXXX
SJC 2603	2603	SJC-Everywhere-CSS	Ian Gillan	1408555XXXX
SJC 2604	2604	SJC-Everywhere-CSS	Keith Emerson	1408555XXXX
SJC 2605	2605	SJC-Everywhere-CSS	John Lennon	1408555XXXX

Enter values 8

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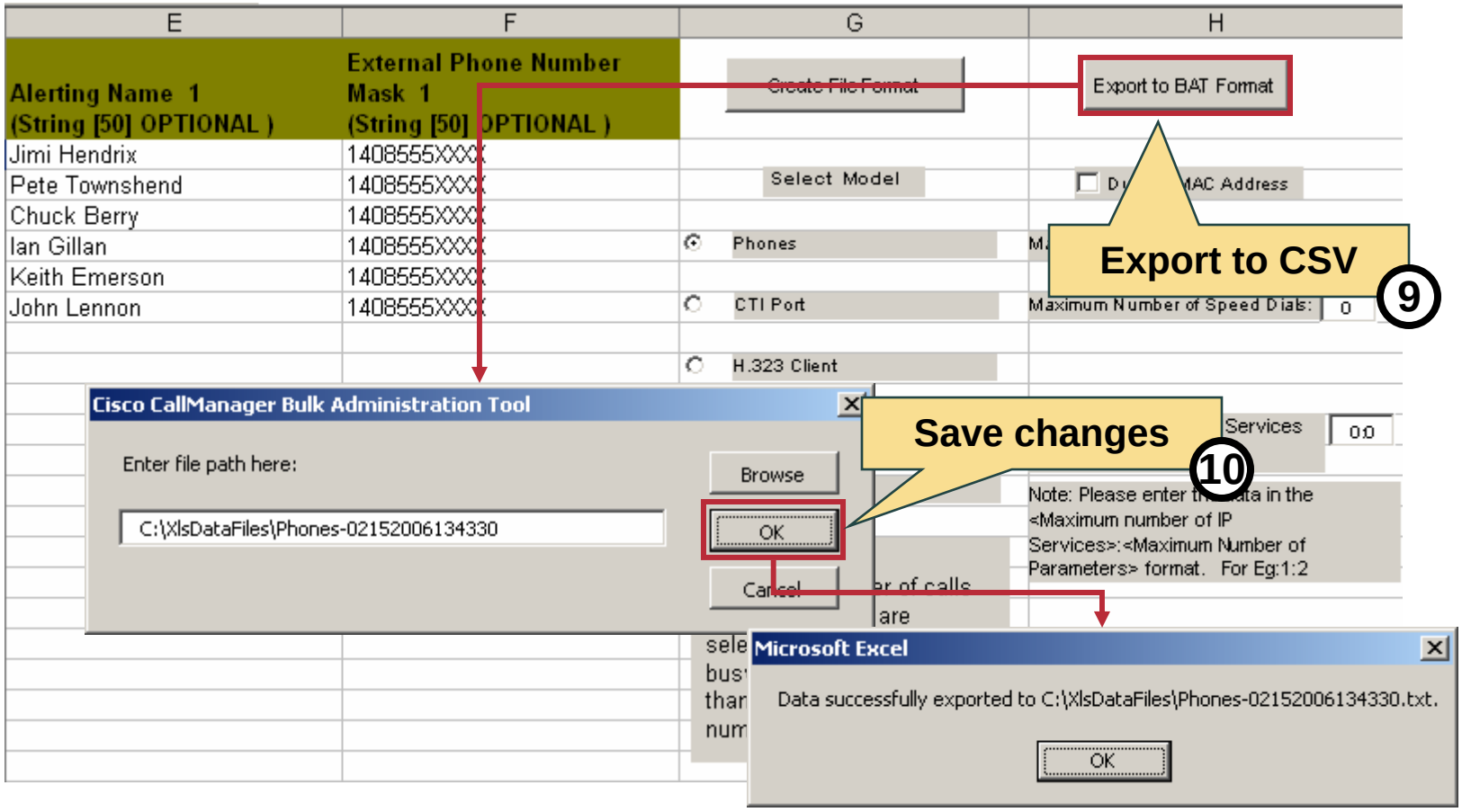
Creating CSV Files



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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.**

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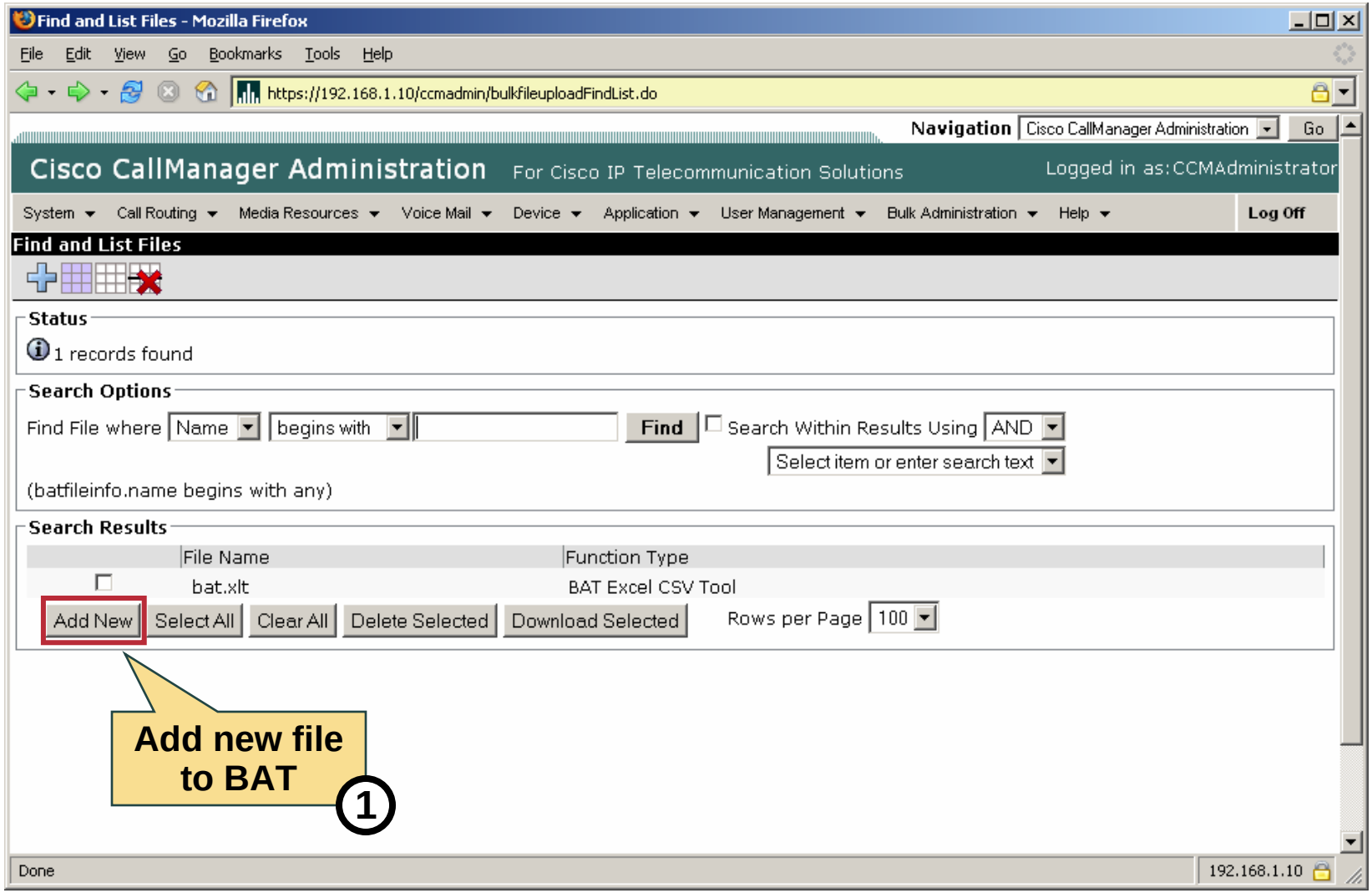
Adding Devices



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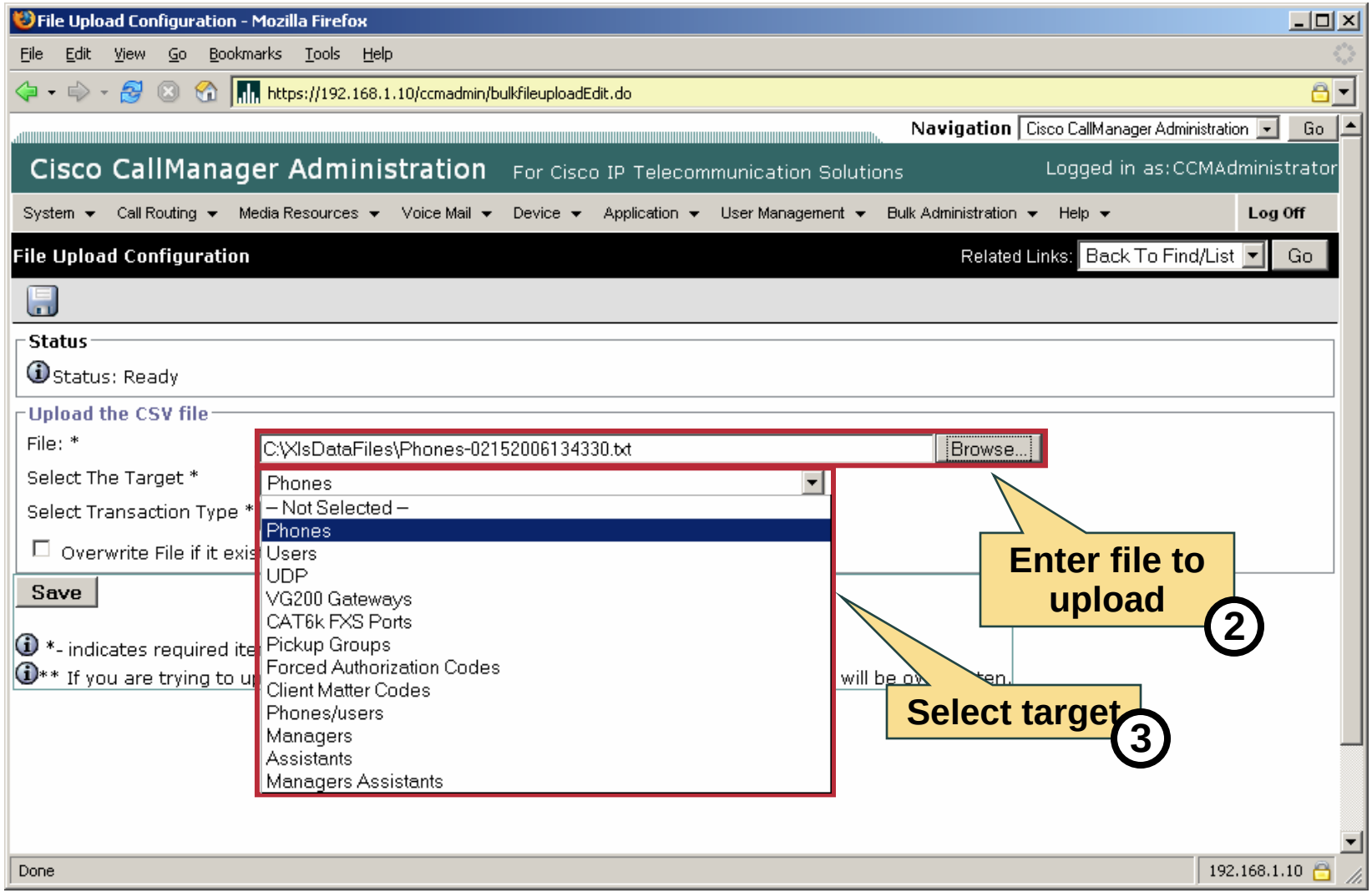
Adding Devices



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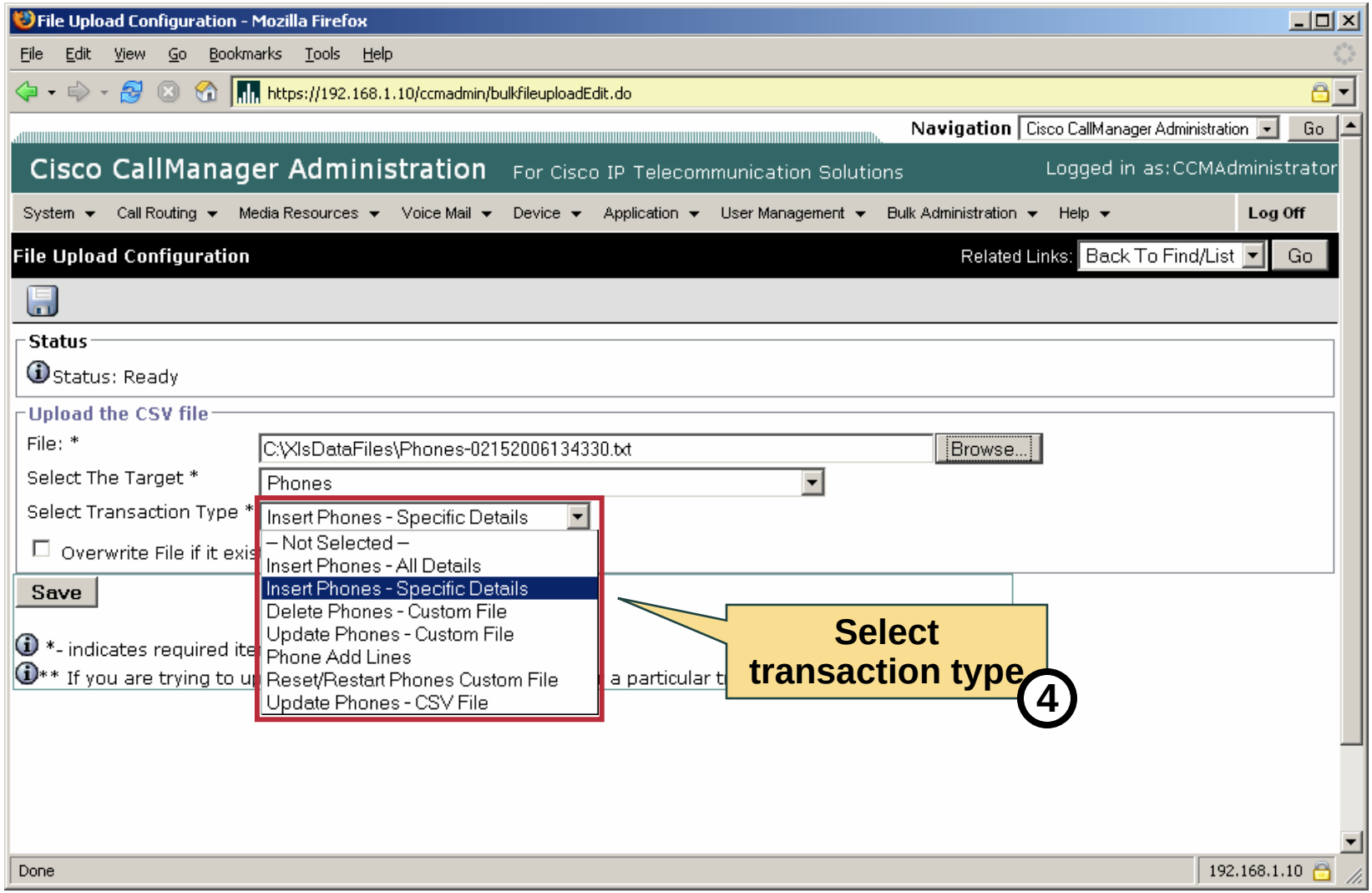
Adding Devices



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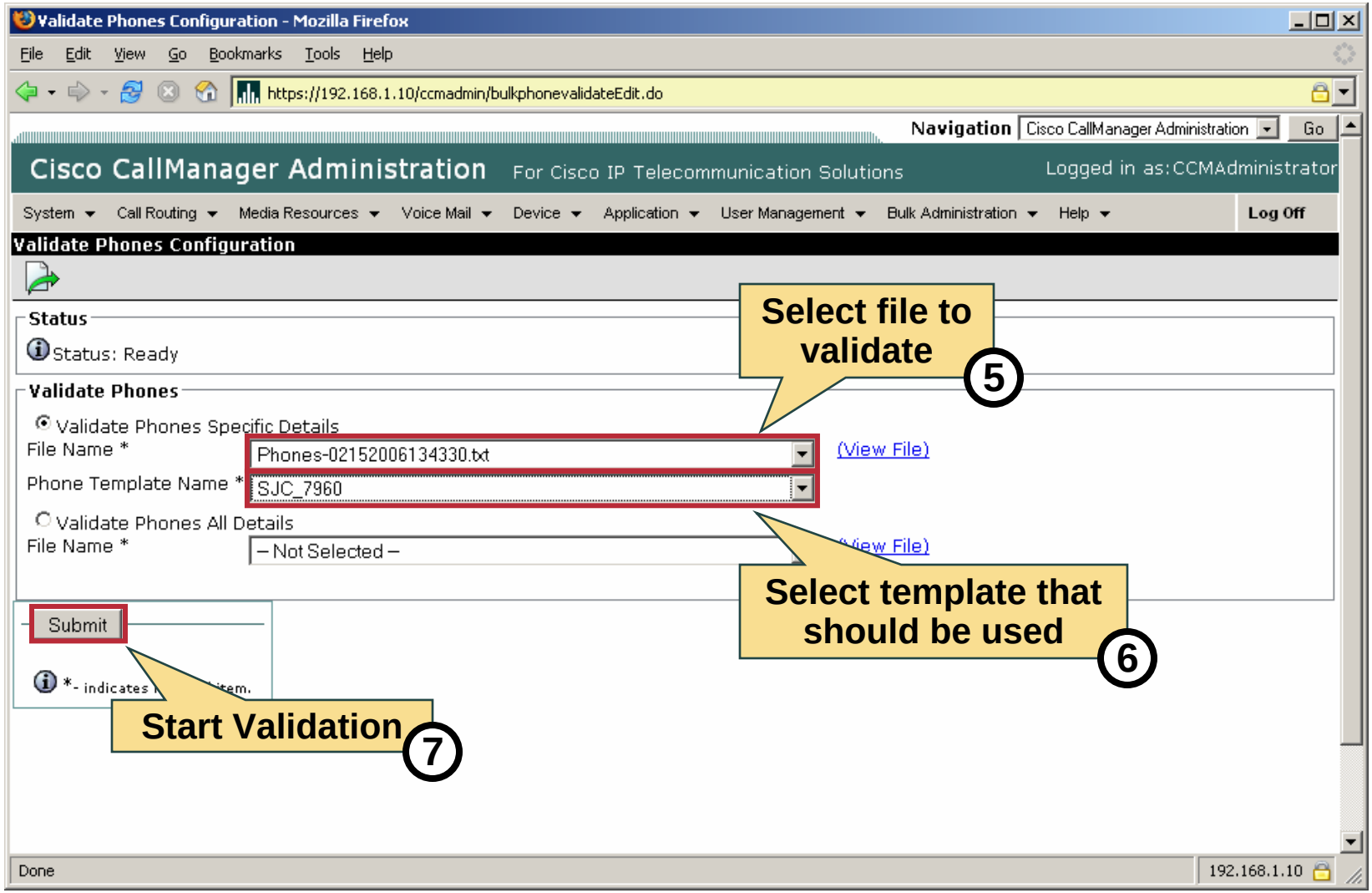
Adding Devices



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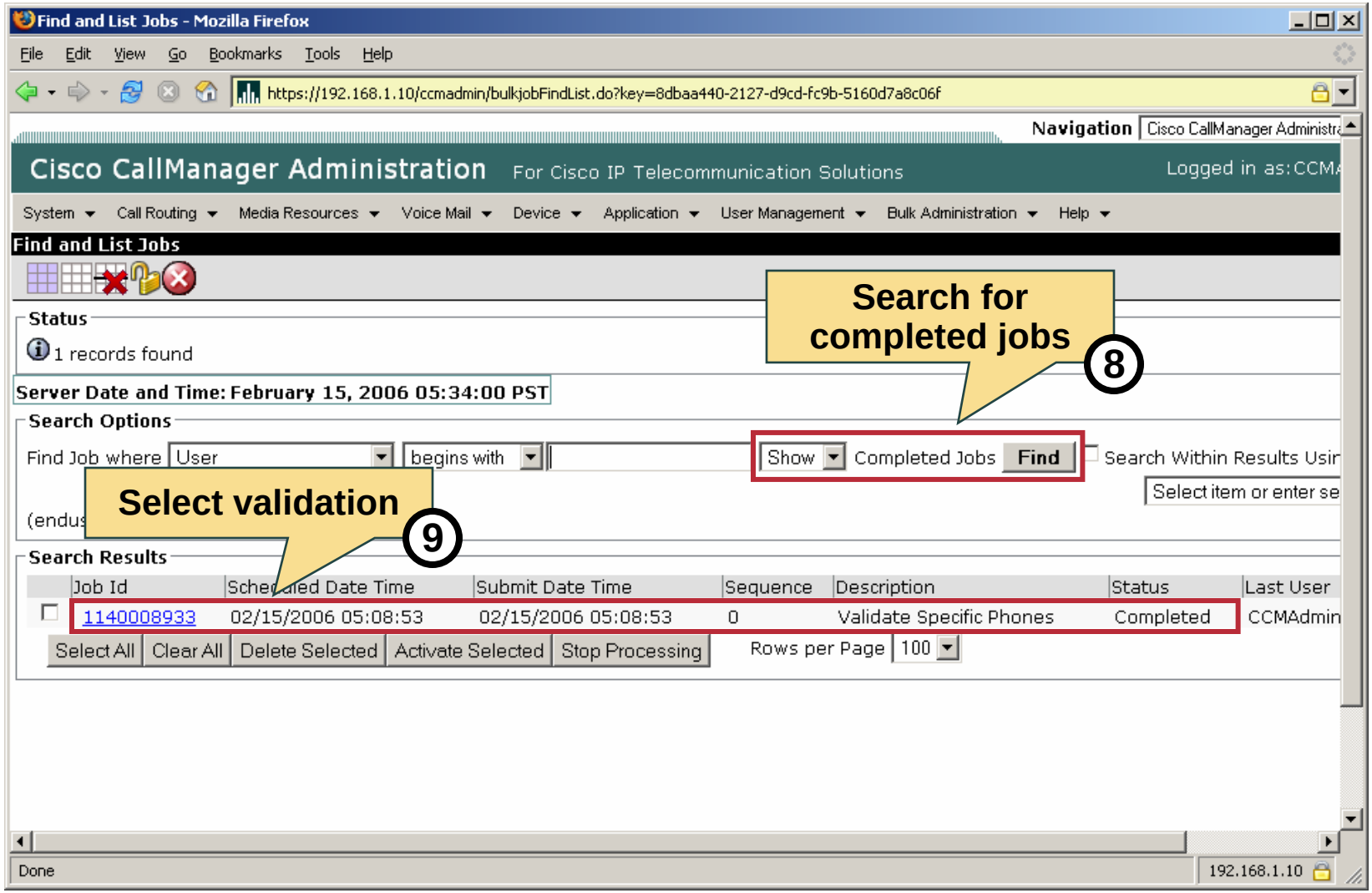
Adding Devices



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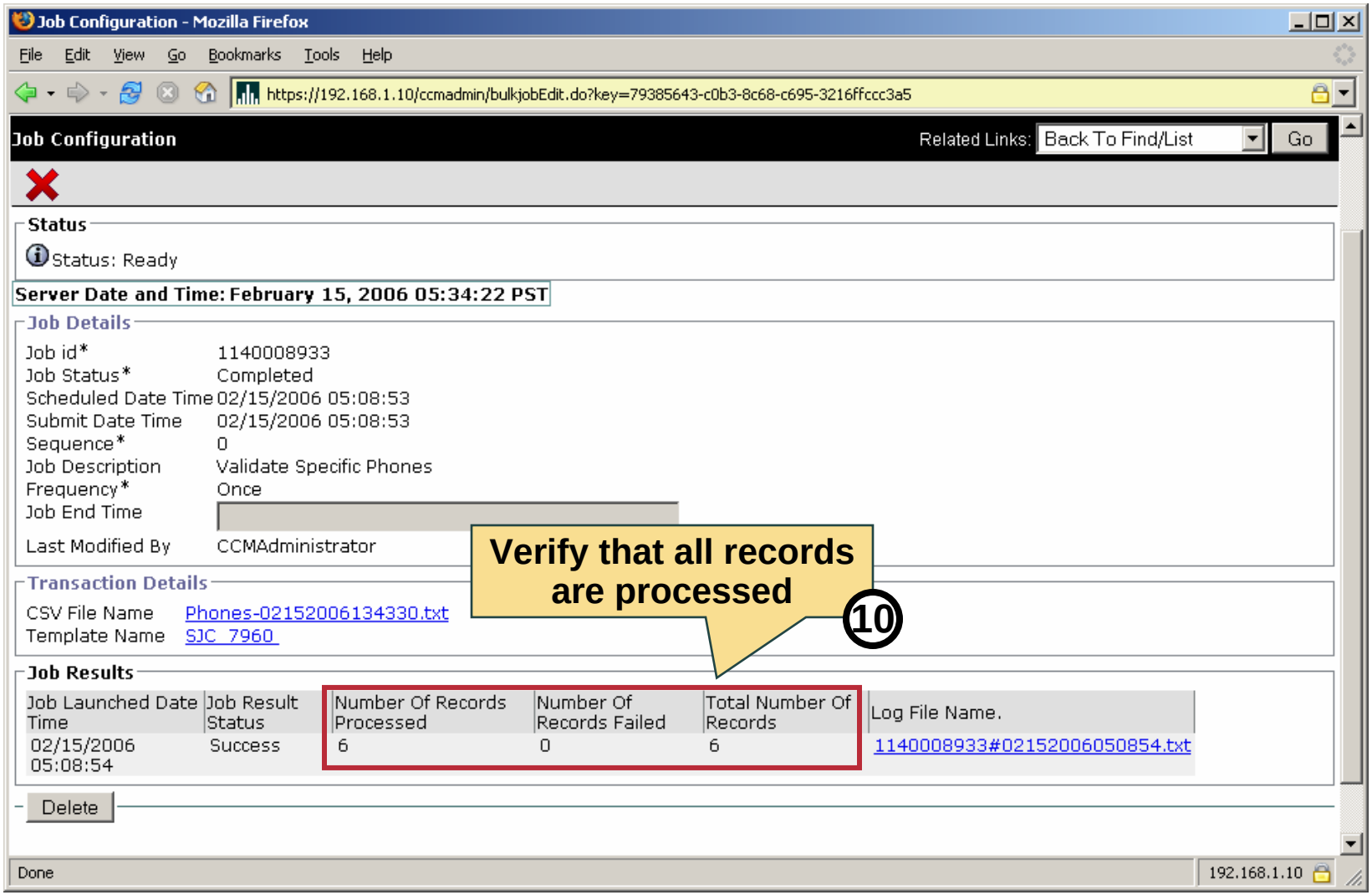
Adding Devices



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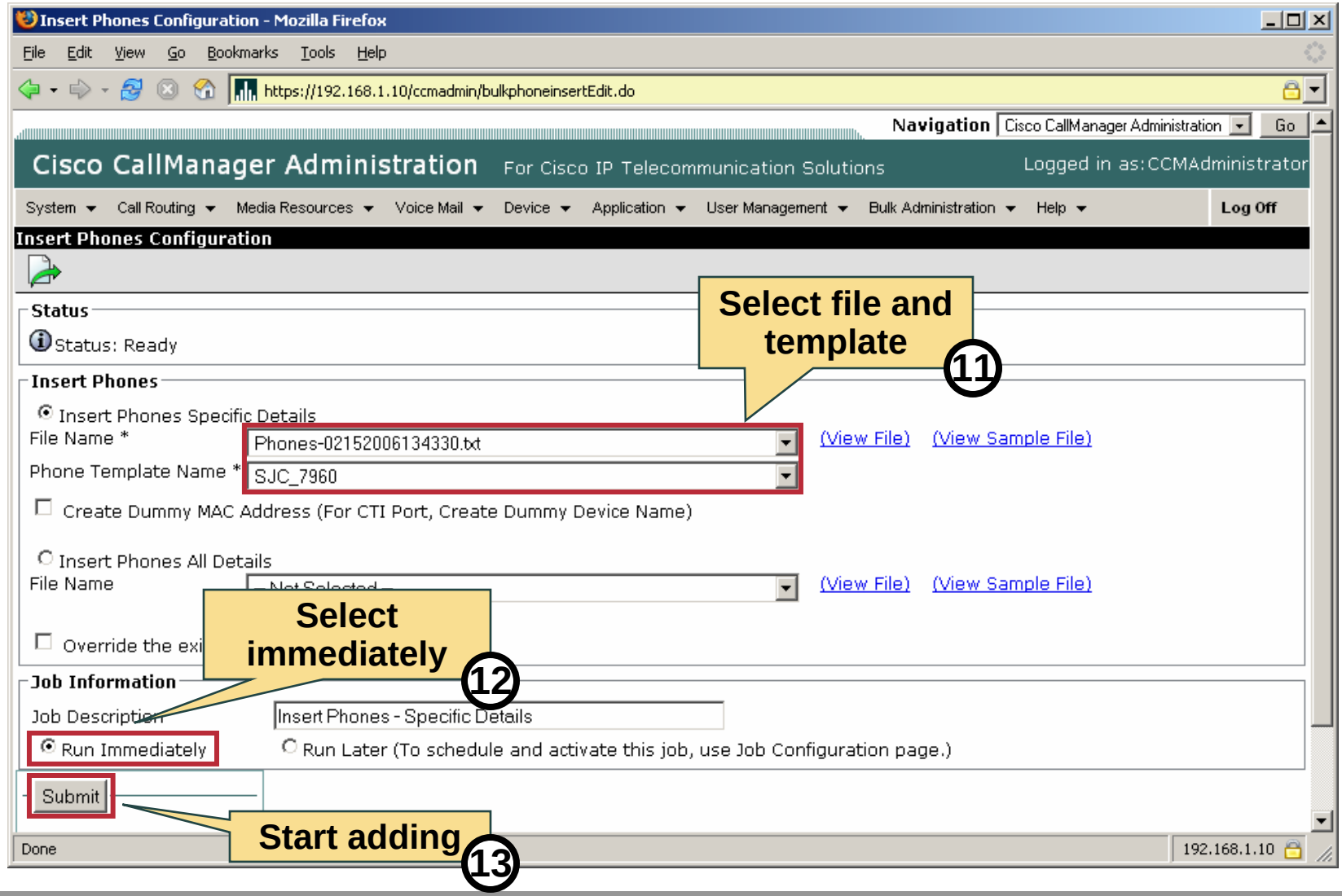
Adding Devices



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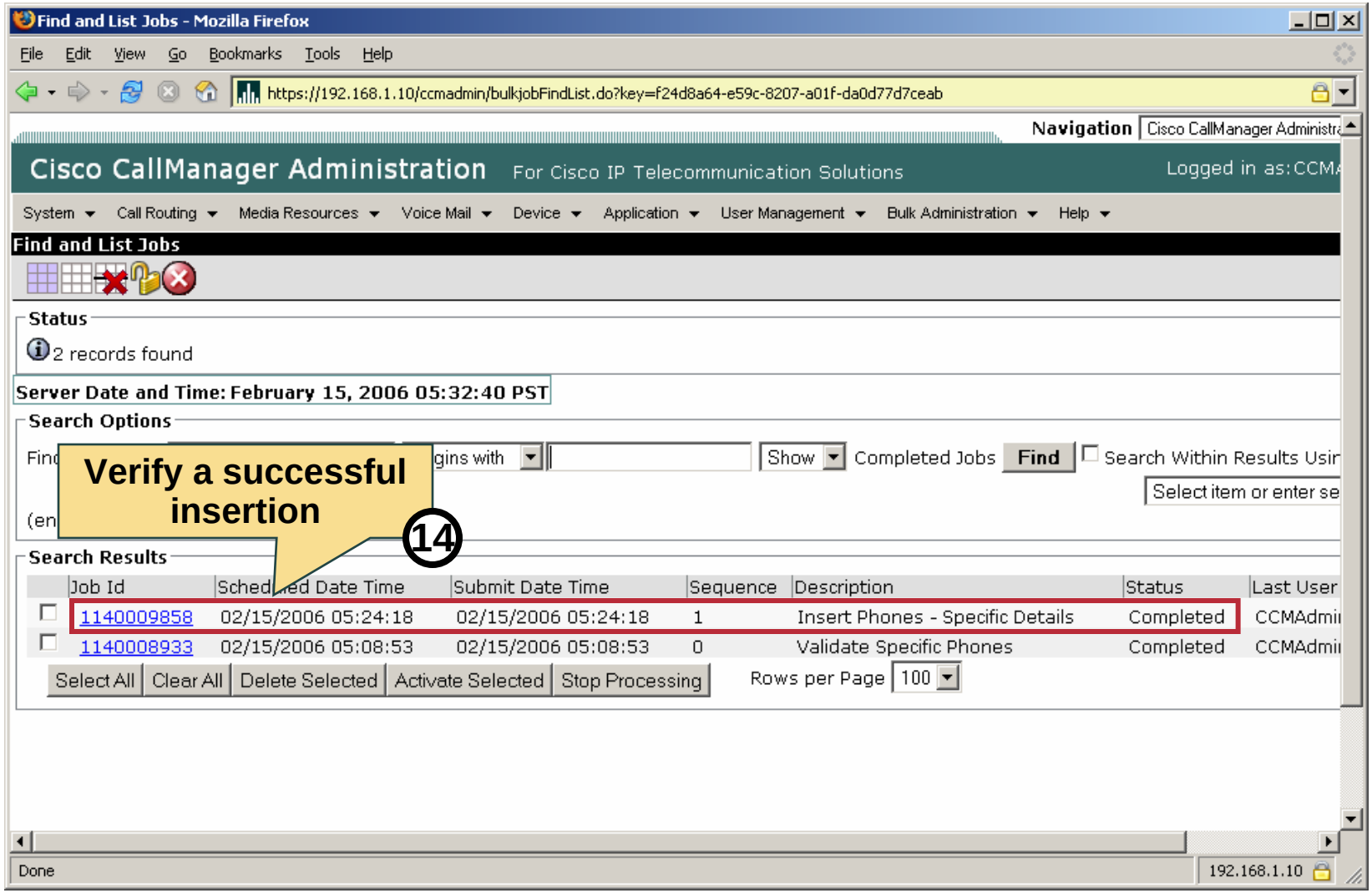
Adding Devices



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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.**

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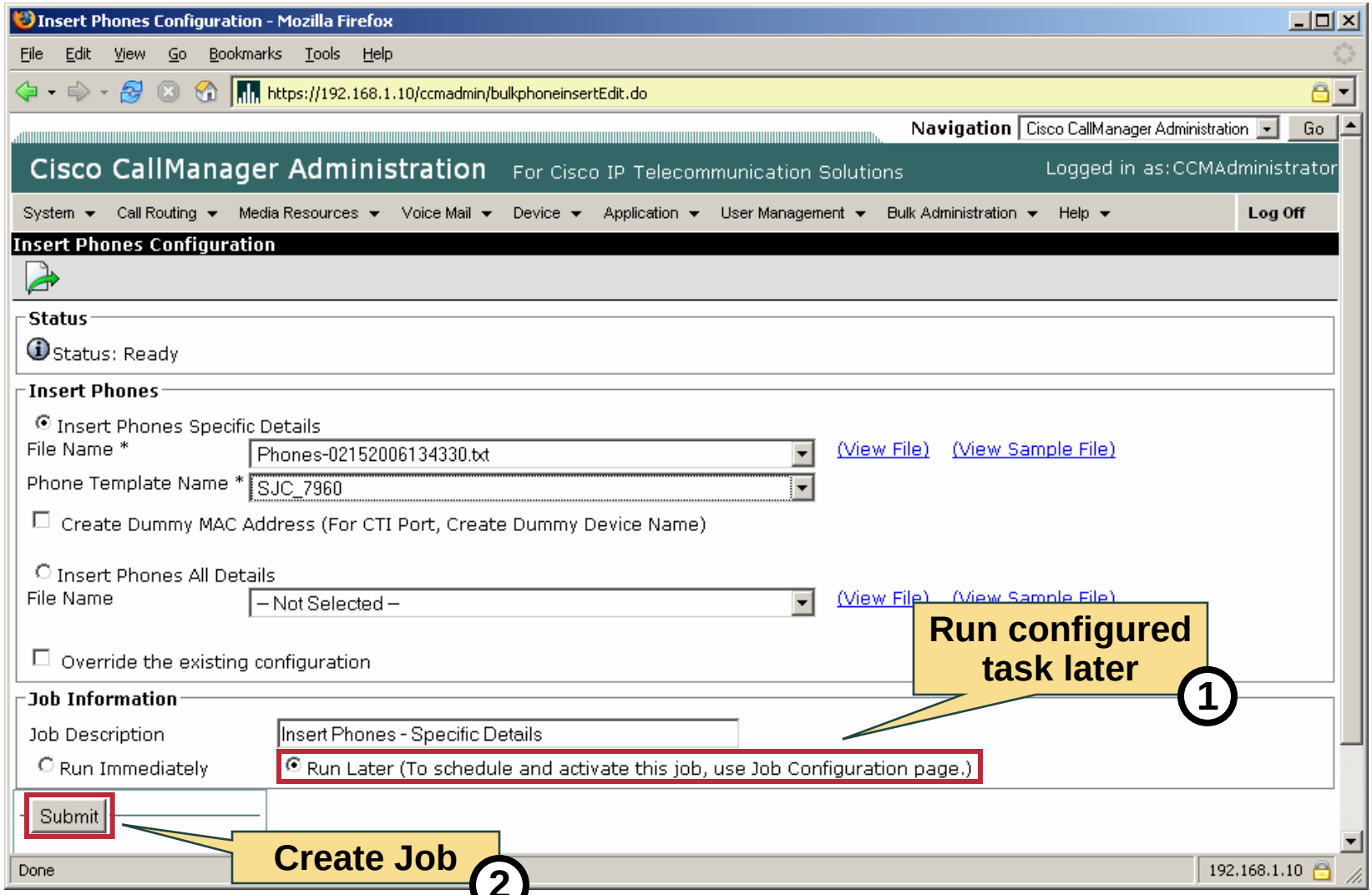
Scheduling Tasks



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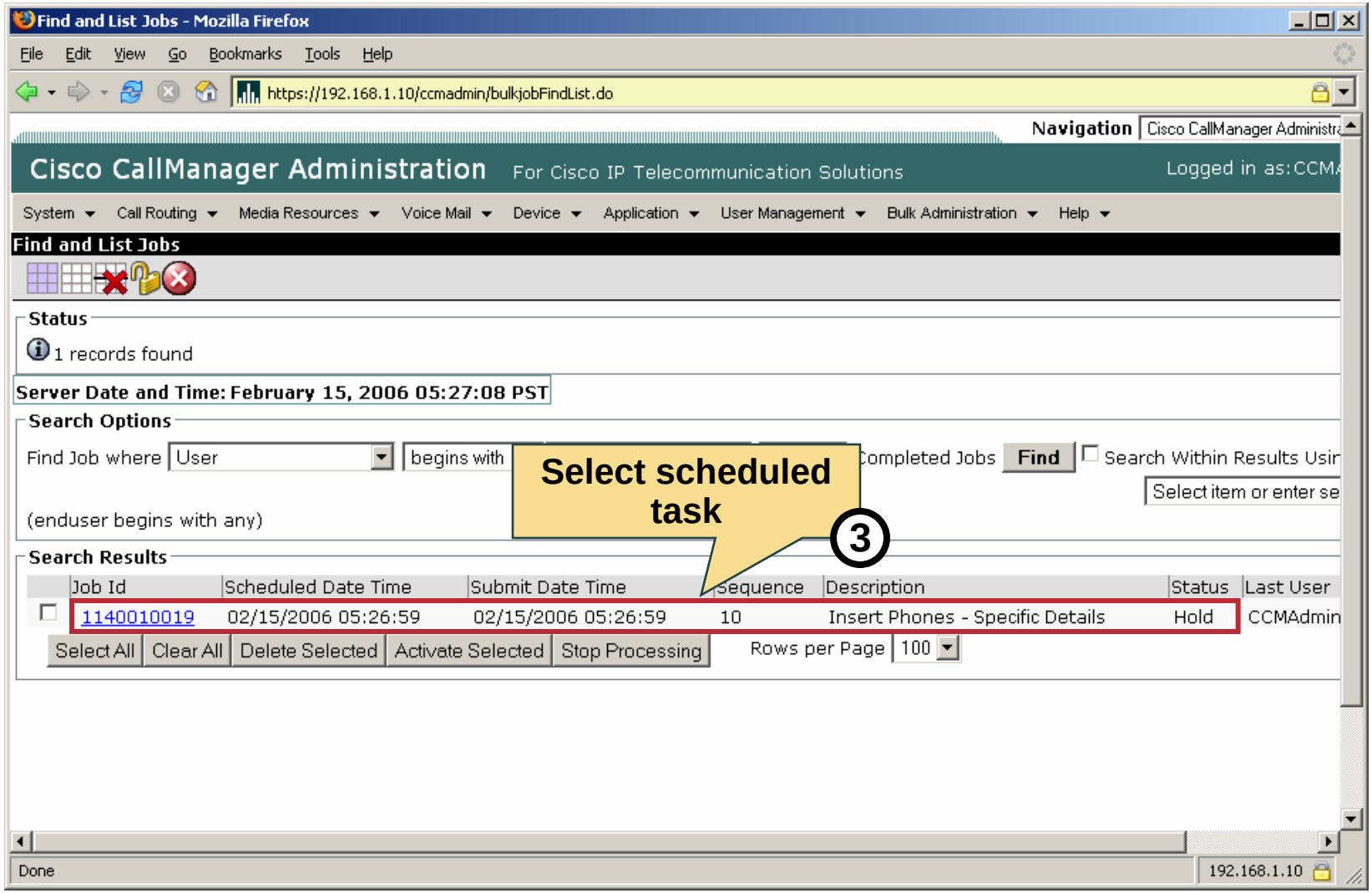
Scheduling Tasks



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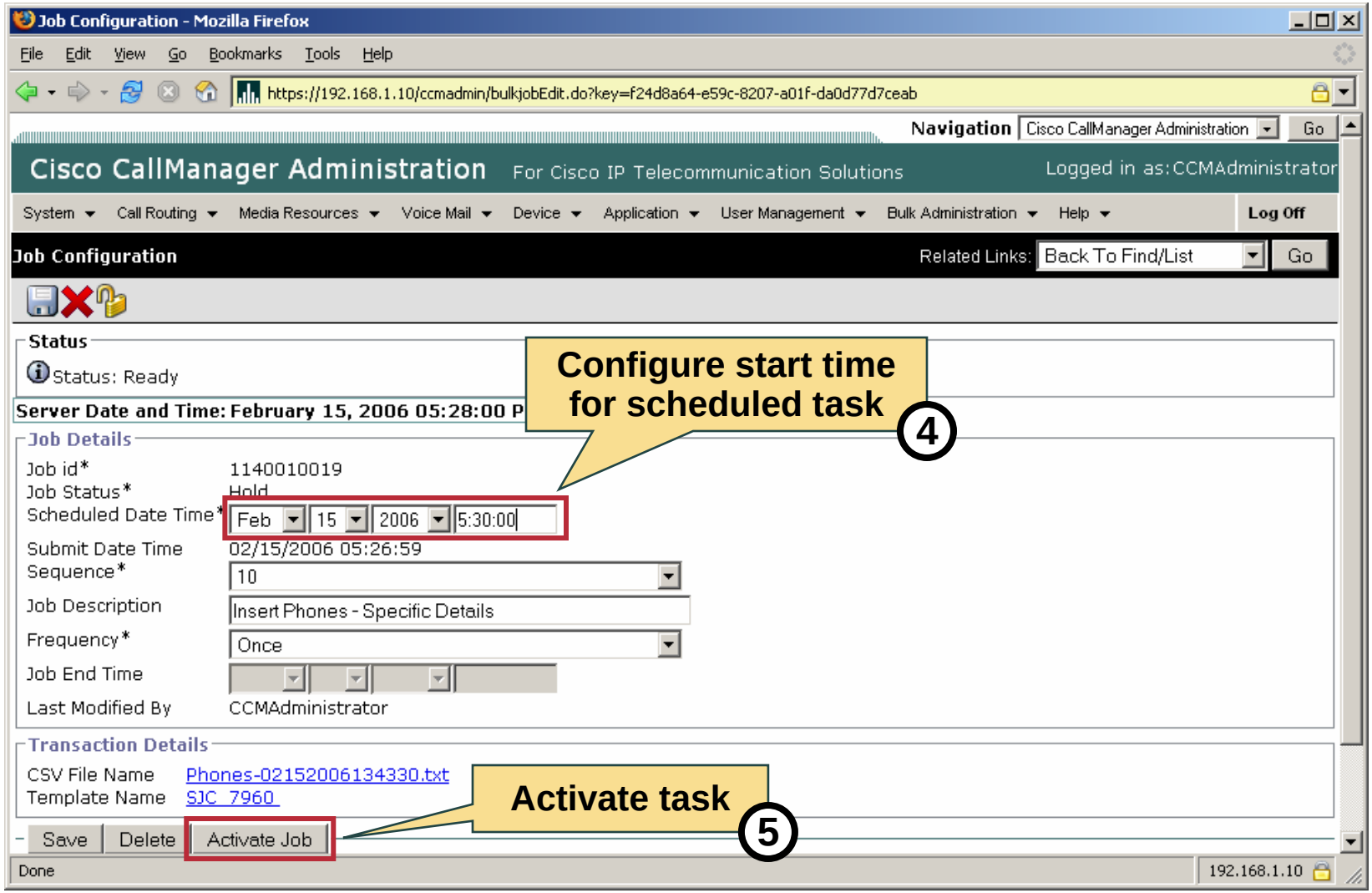
Scheduling Tasks



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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.**

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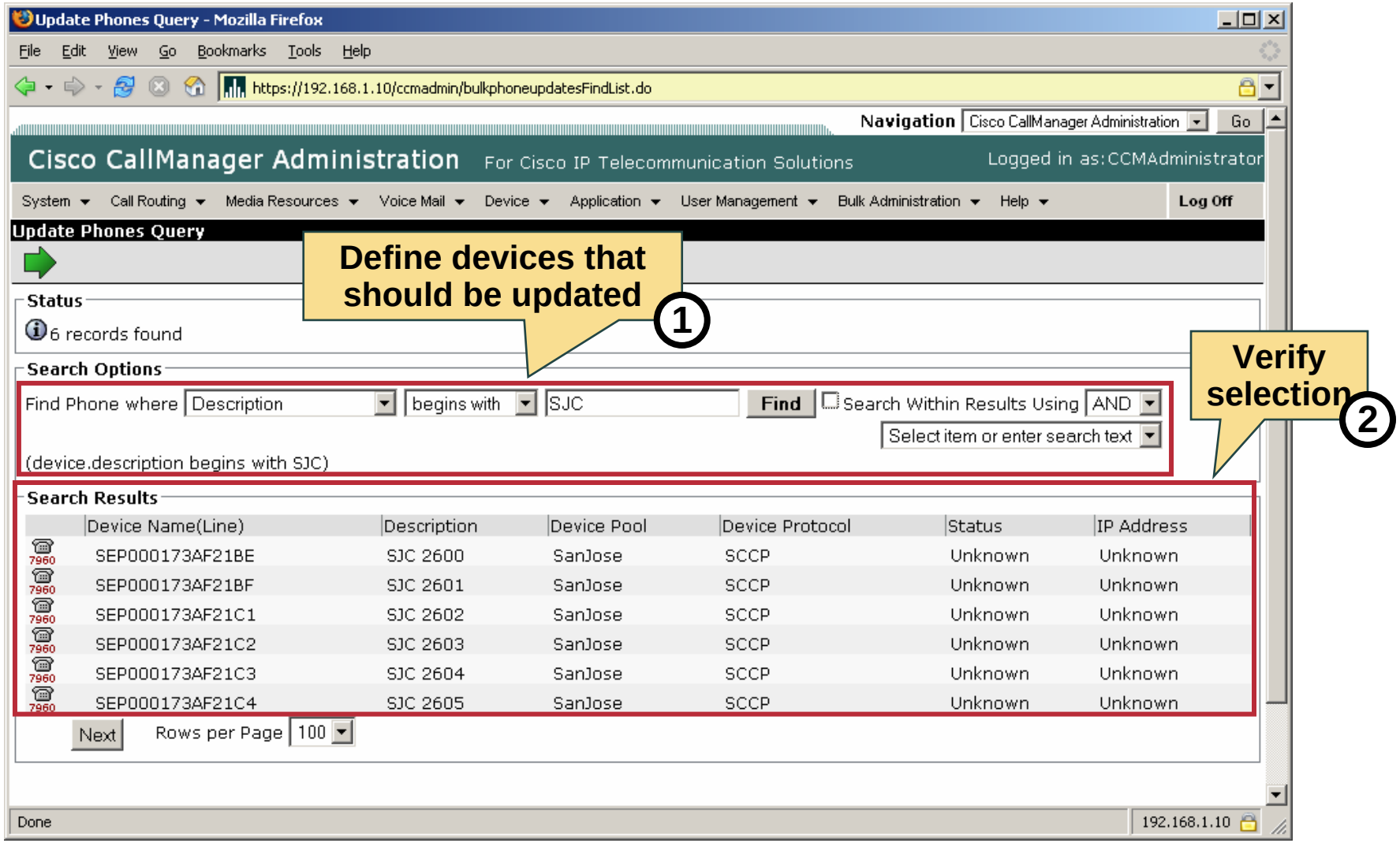
Updating Devices



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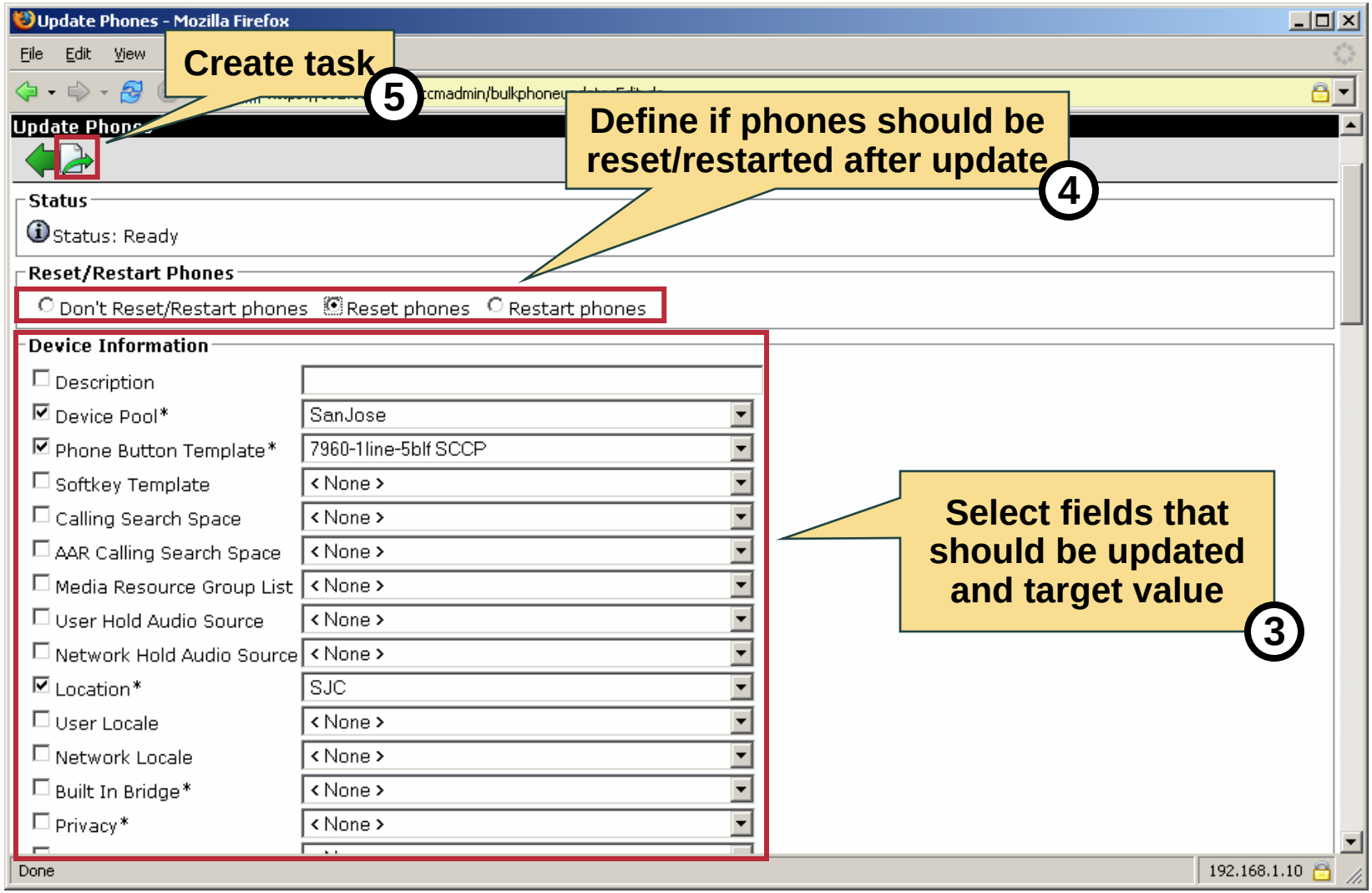
Updating Devices



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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.

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Deploying a Dial Plan on CallManager 5.0

Understanding Dial Plans

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Objectives

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

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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.**

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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.**

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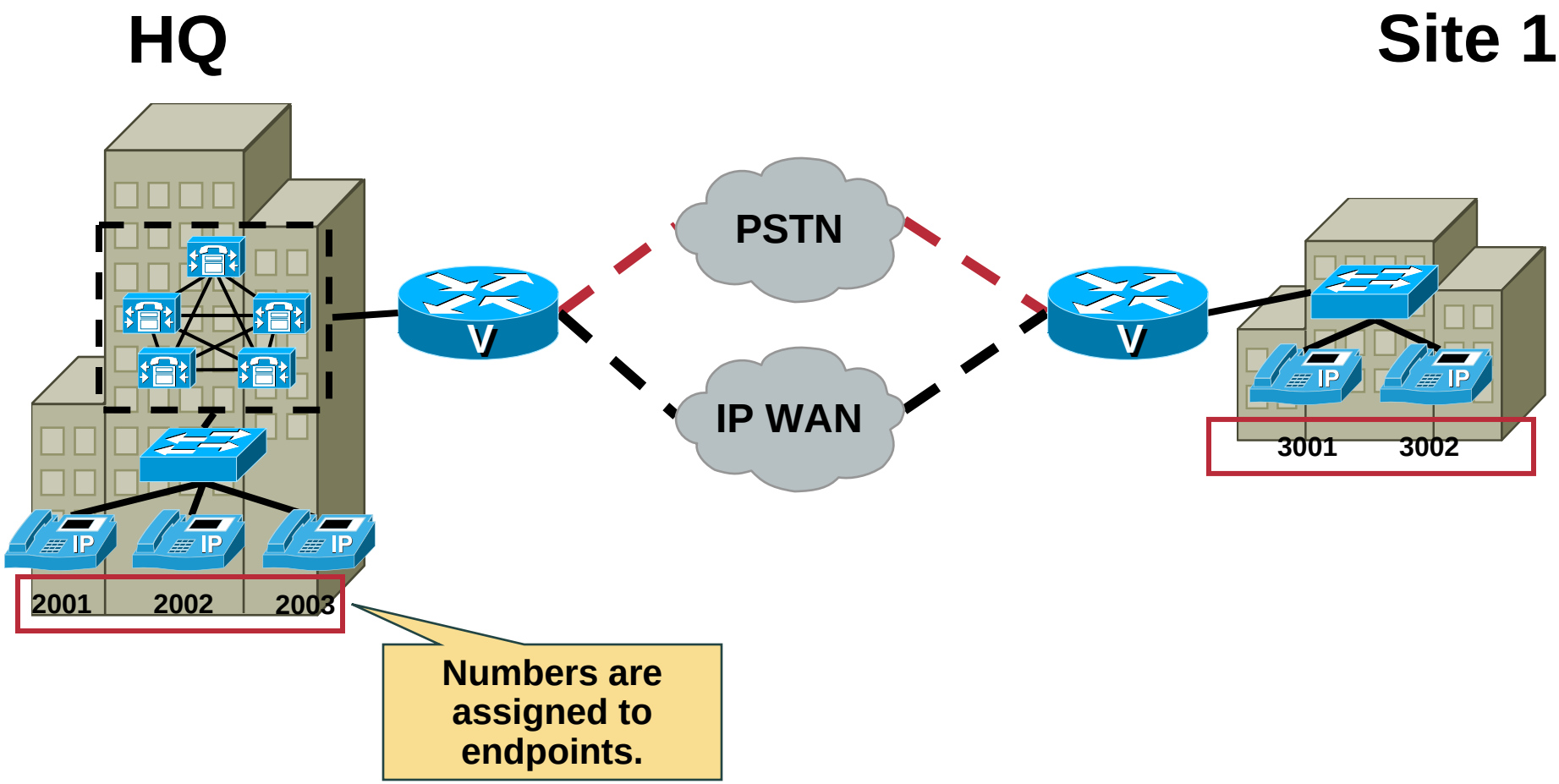
End Point Addressing



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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.**

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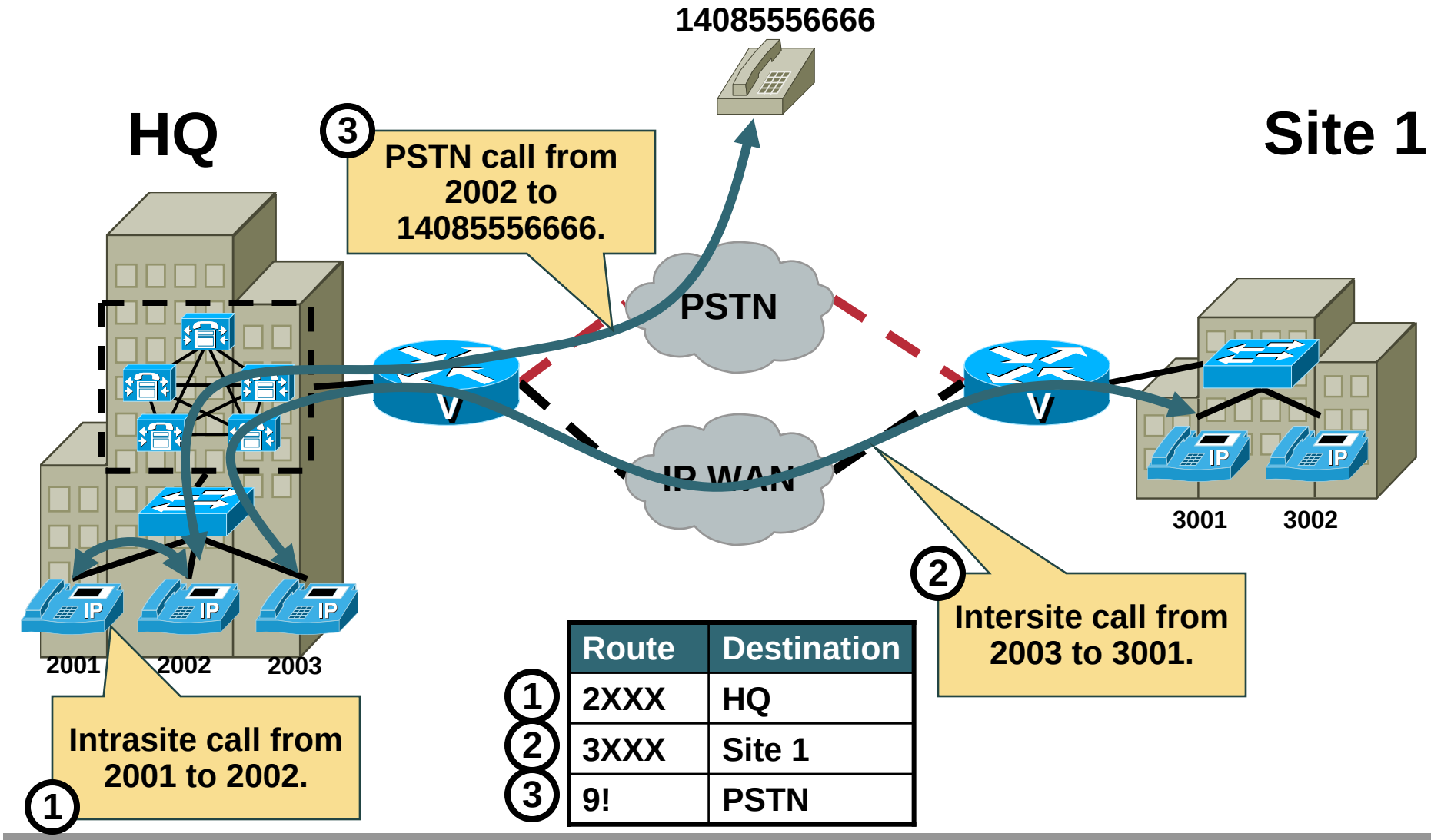
Call Routing and Path Selection



	Route	Destination
1	2XXX	HQ
2	3XXX	Site 1
3	9!	PSTN

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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.**

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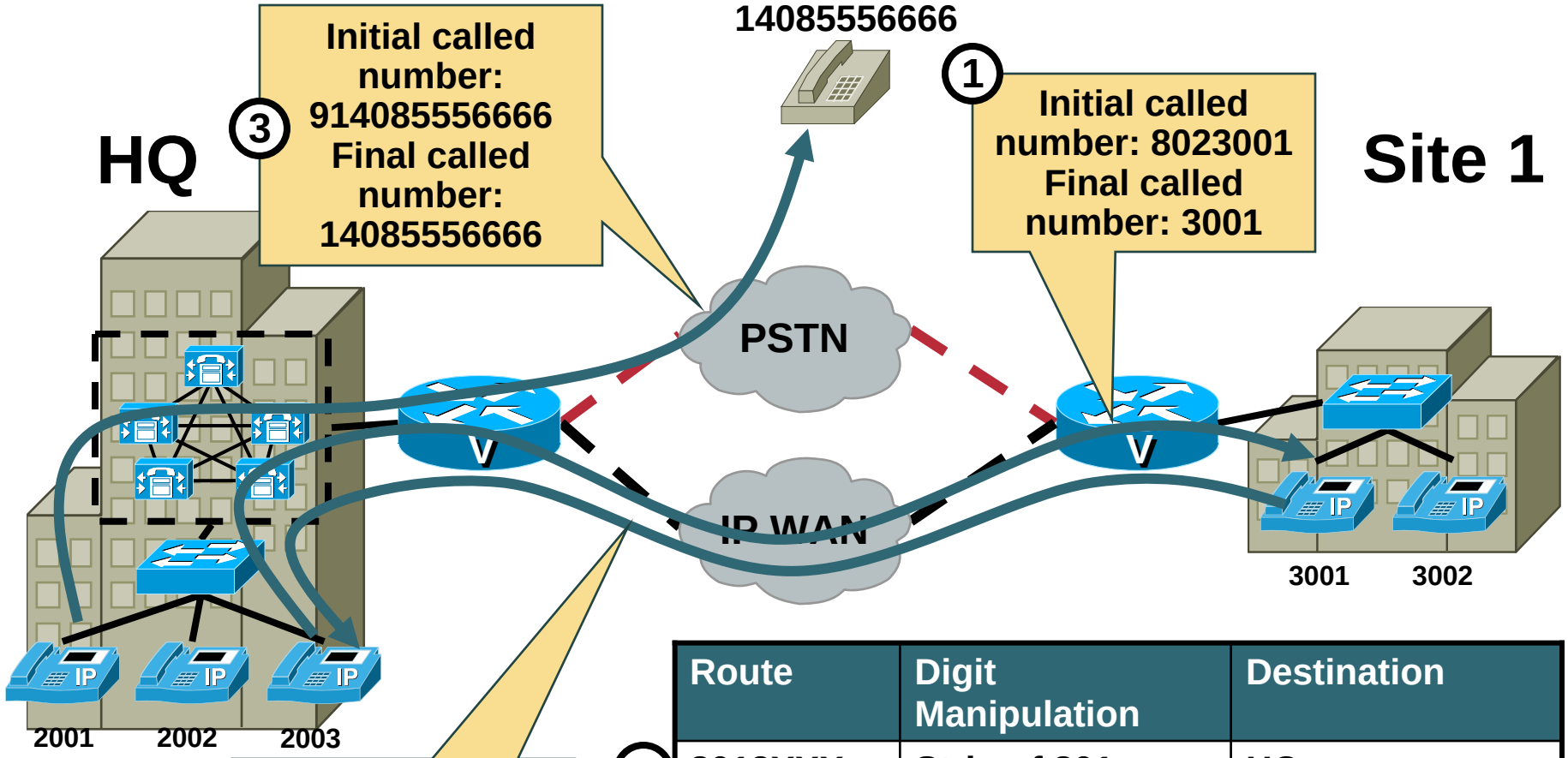
Digit Manipulation



	Route	Digit Manipulation	Destination
1	8012XXX	Strip of 801	HQ
2	8023XXX	Strip of 802	Site 1
3	9!	Strip of 9	PSTN

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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.**

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Call Privileges

Calling Privilege Class (Class of Service)	Allowed Destinations
Internal	• Internal • Emergency
Local	• Internal • Emergency • Local PSTN
Long Distance	• Internal • Emergency • Local PSTN • Long Distance PSTN
International	• Internal • Emergency • Local PSTN • Long Distance PSTN • International PSTN

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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**

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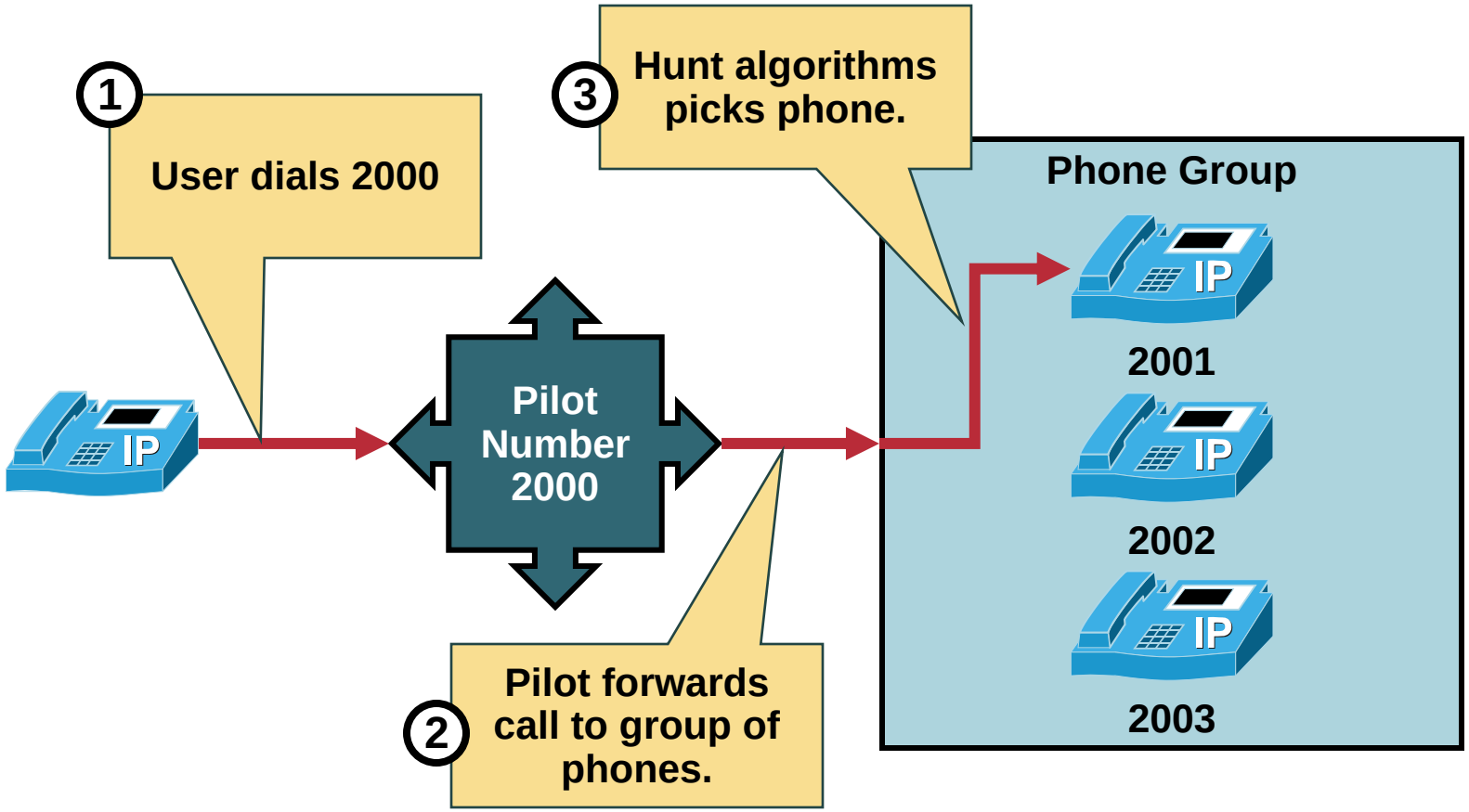
Call Coverage



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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.

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Deploying a Dial Plan on CallManager 5.0

Understanding Cisco Unified CallManager Dial Plan Components

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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

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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**

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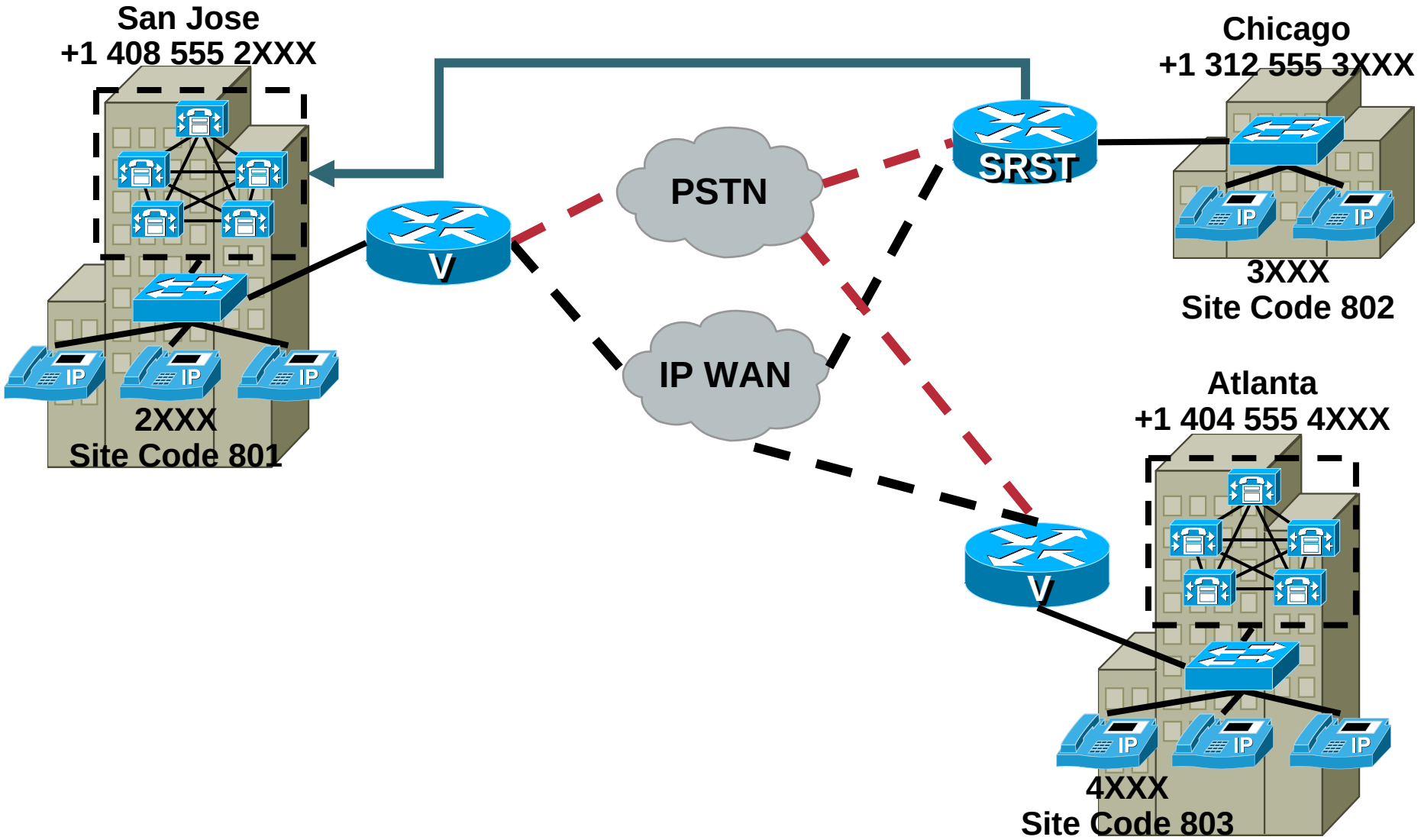
Cisco Unified CallManager Dial Plan Components Overview



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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**

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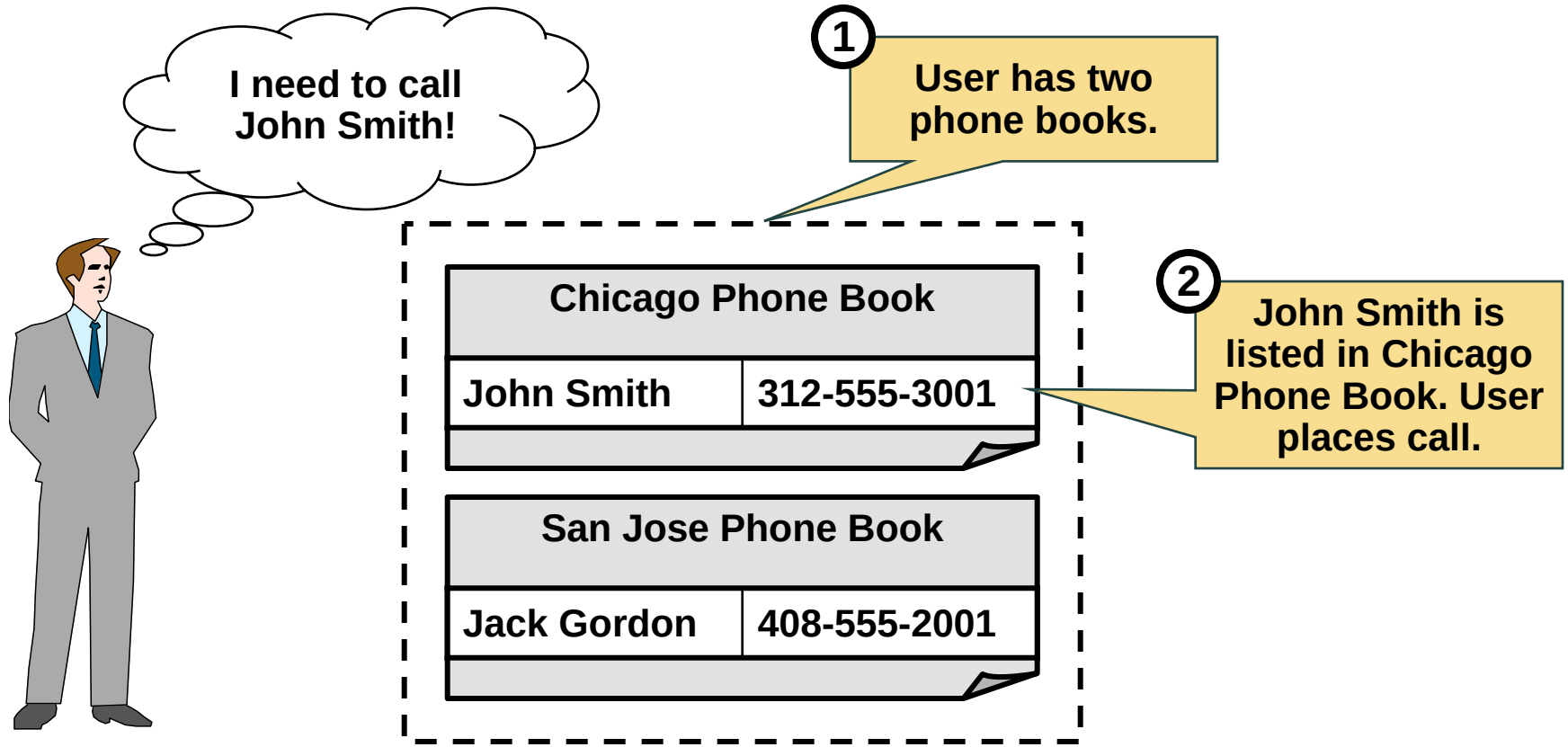
Partitions and Calling Search Spaces



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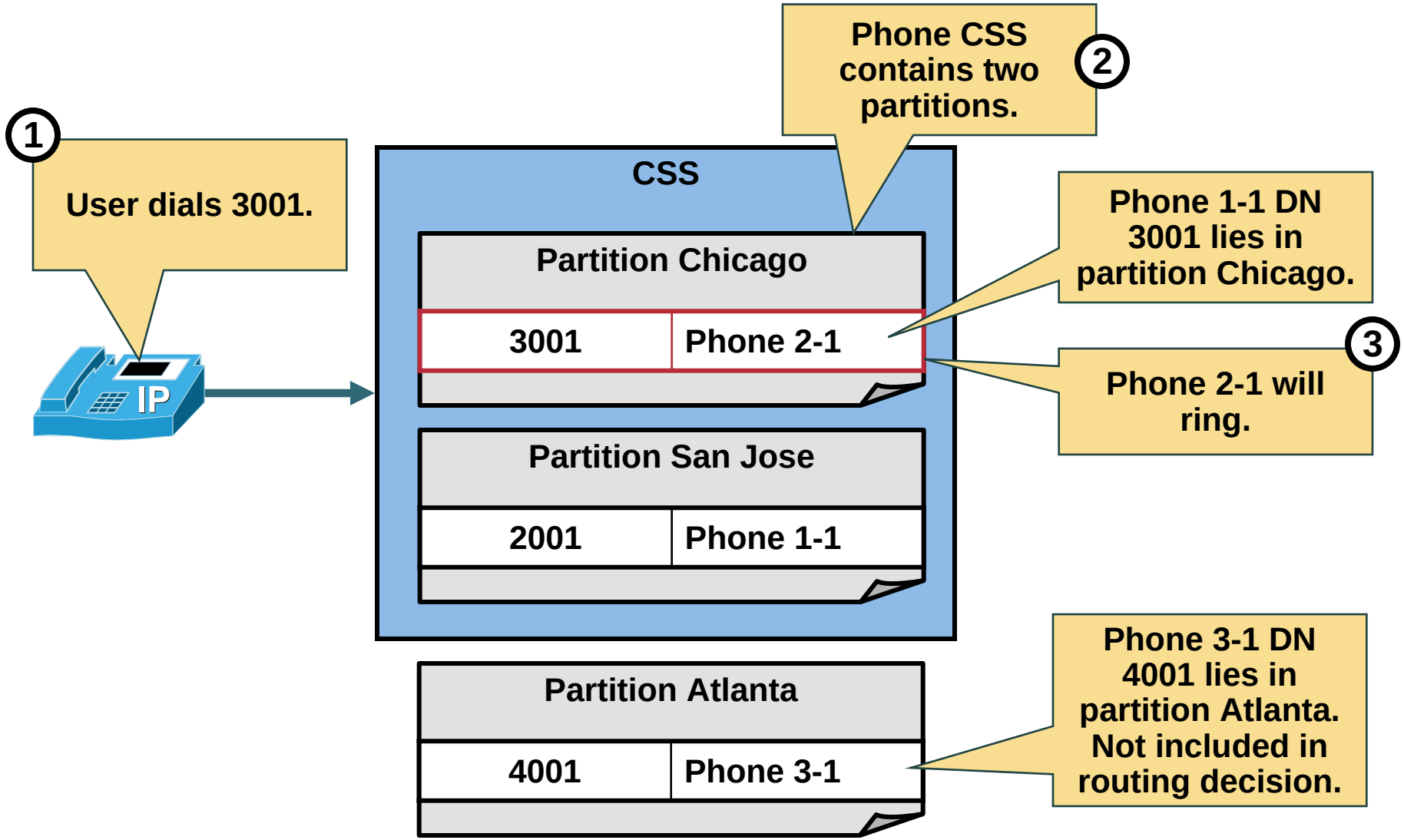
Partitions and Calling Search Spaces



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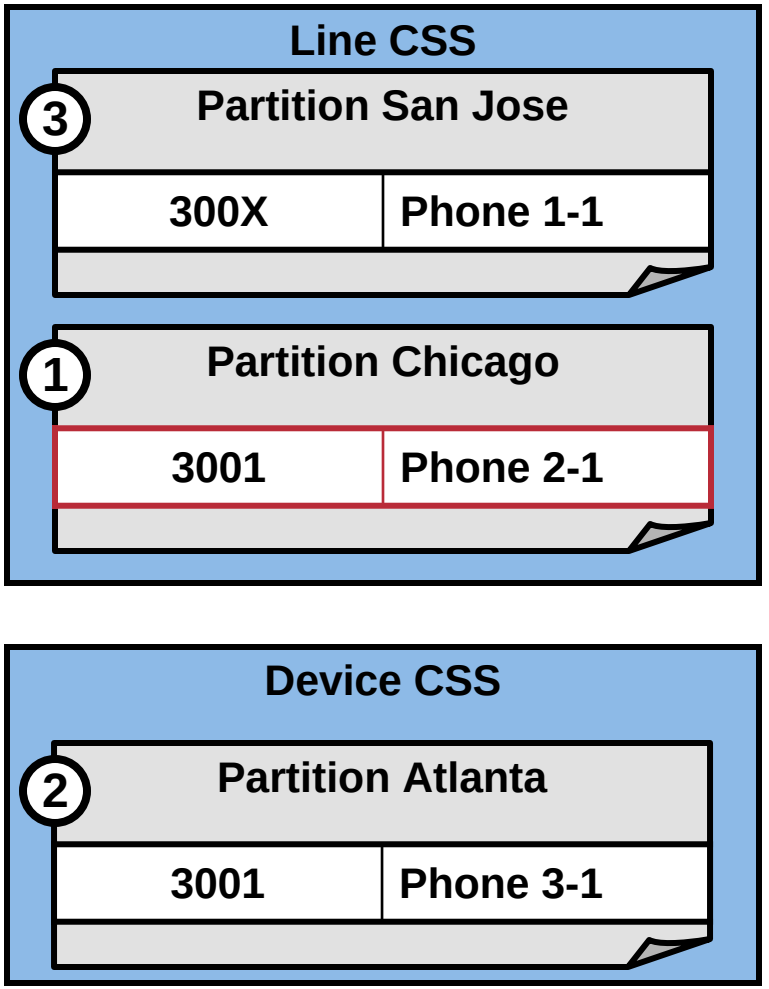
Partitions and Calling Search Spaces

Route selection uses the following rules:

- Closest match routing.
  - Similar to IP routing.
  - X wildcard is less specifc.
- Partition order in CSS only used as a tie breaker.
  - Top = Highest Priority
  - Bottom = Lowest Priority
  - Line CSS has higher priority than phone CSS.



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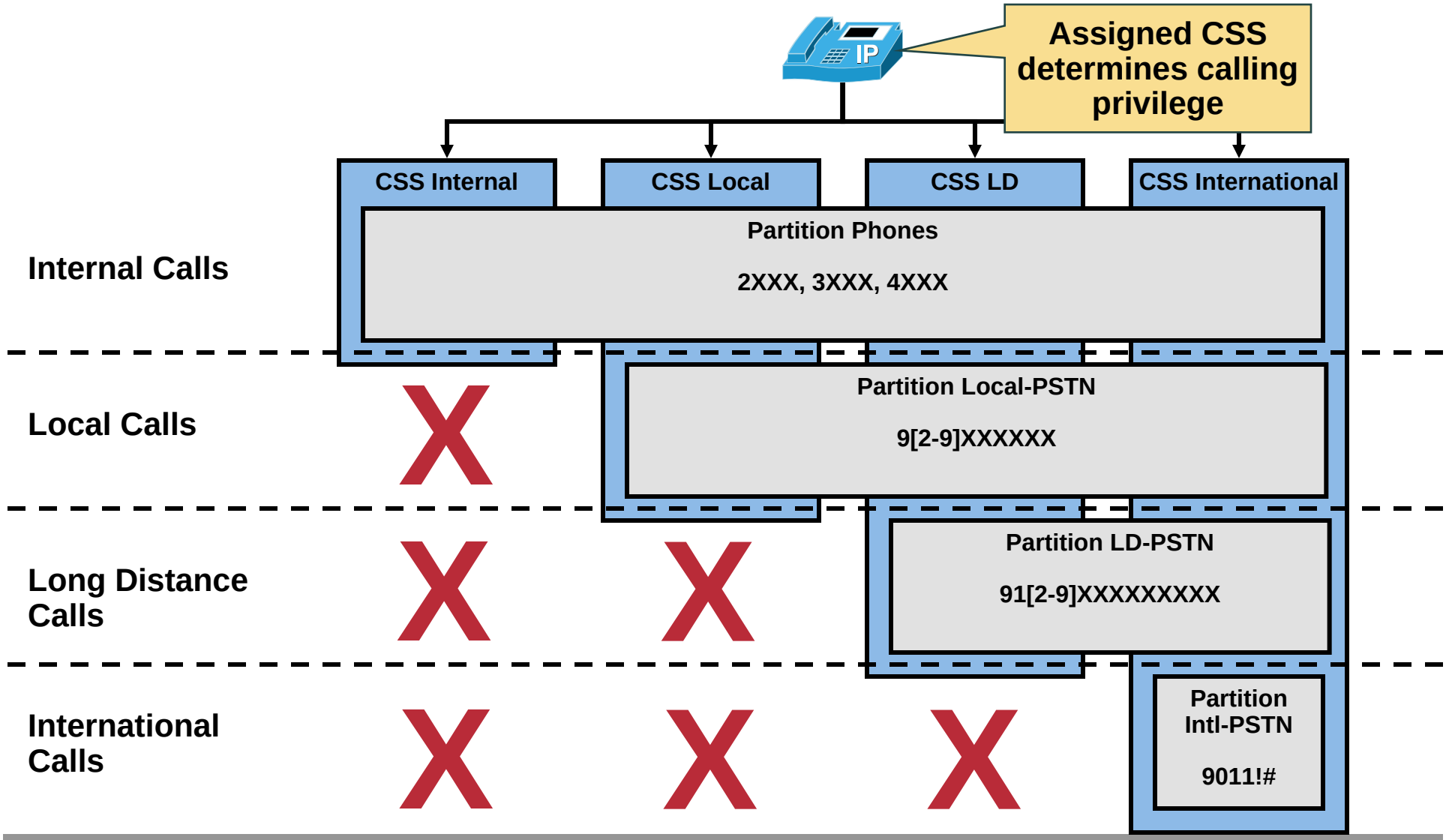
Partitions and Calling Search Spaces



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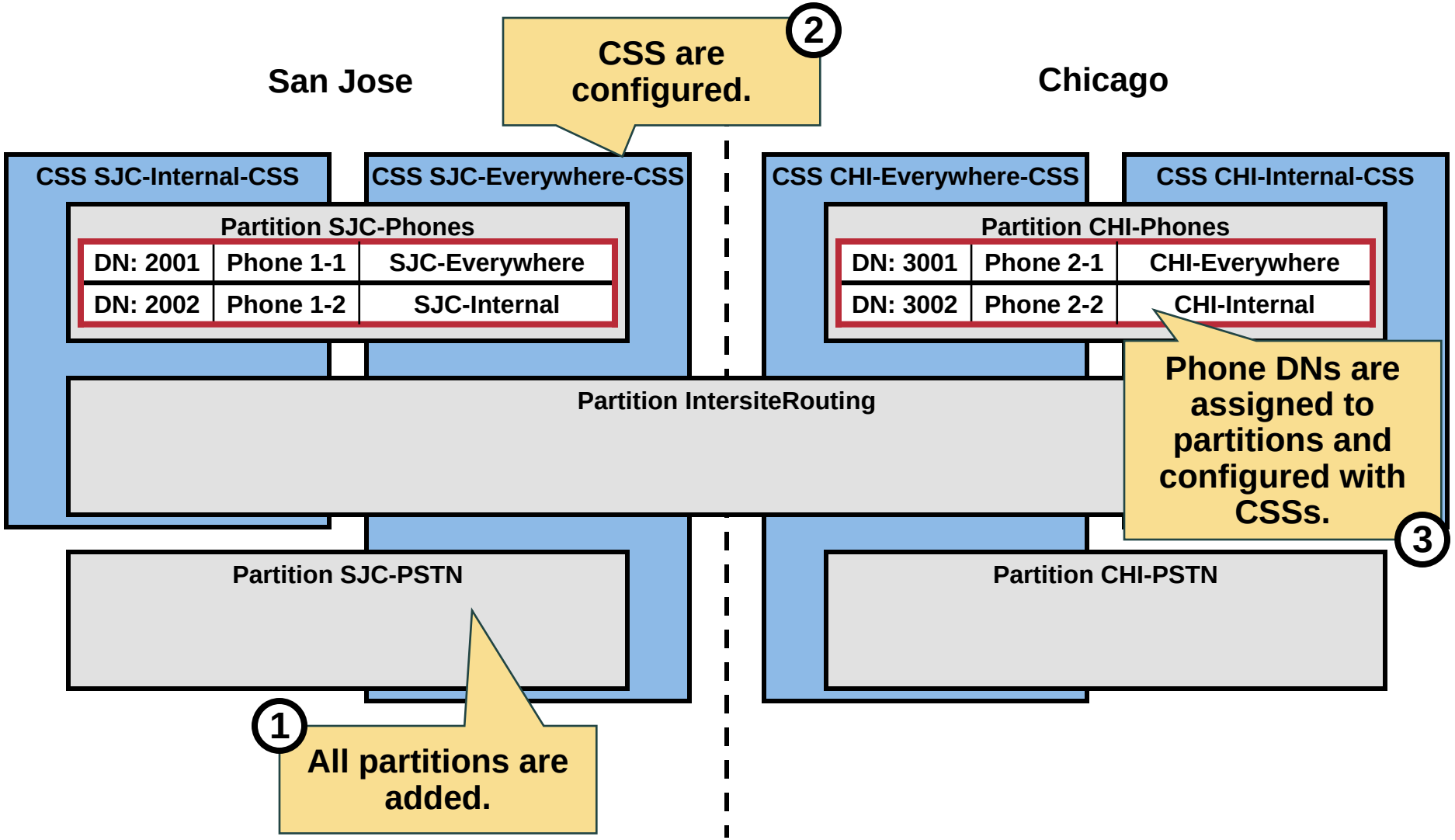
Partitions and Calling Search Spaces



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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.**

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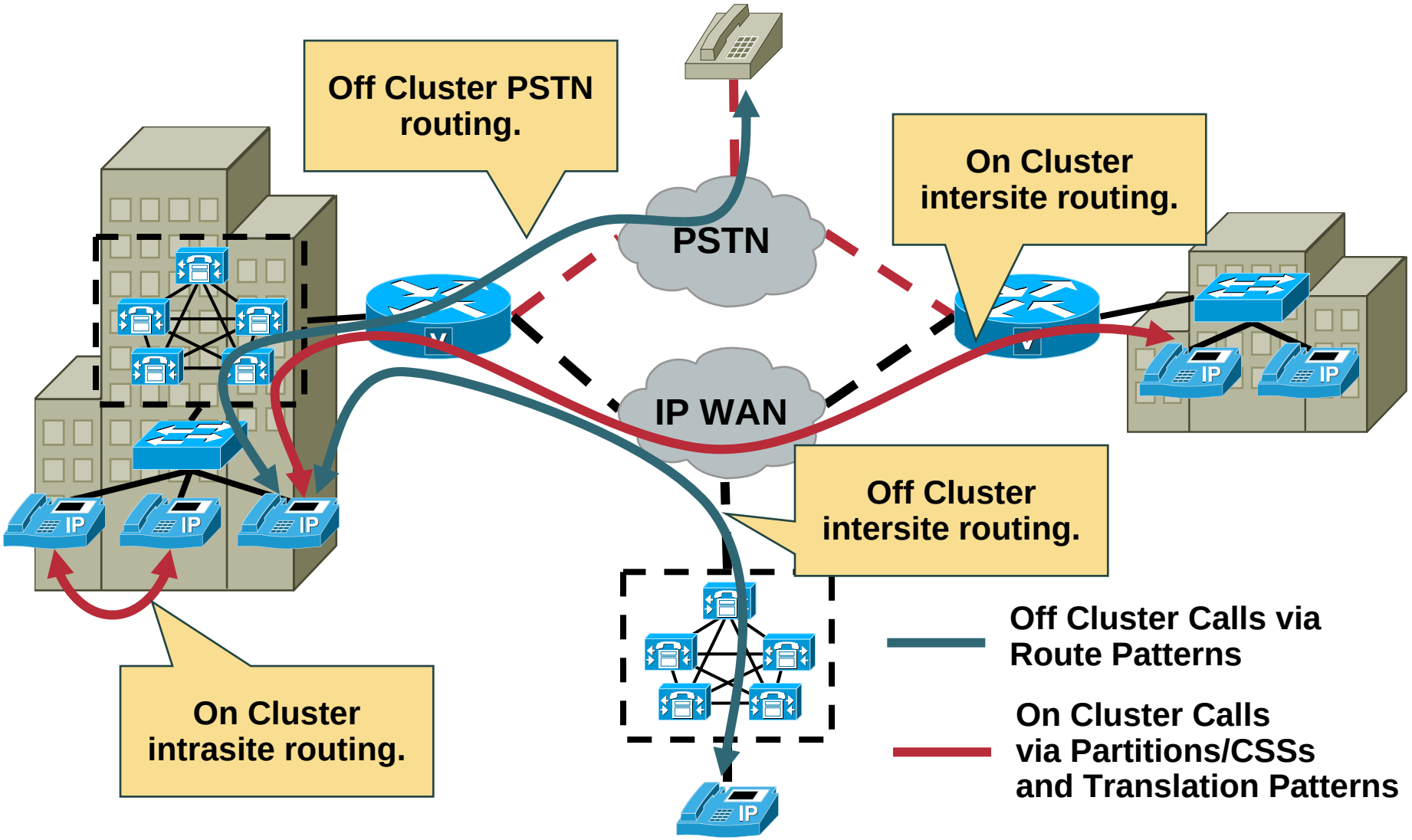
On/Off Cluster Routing



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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**

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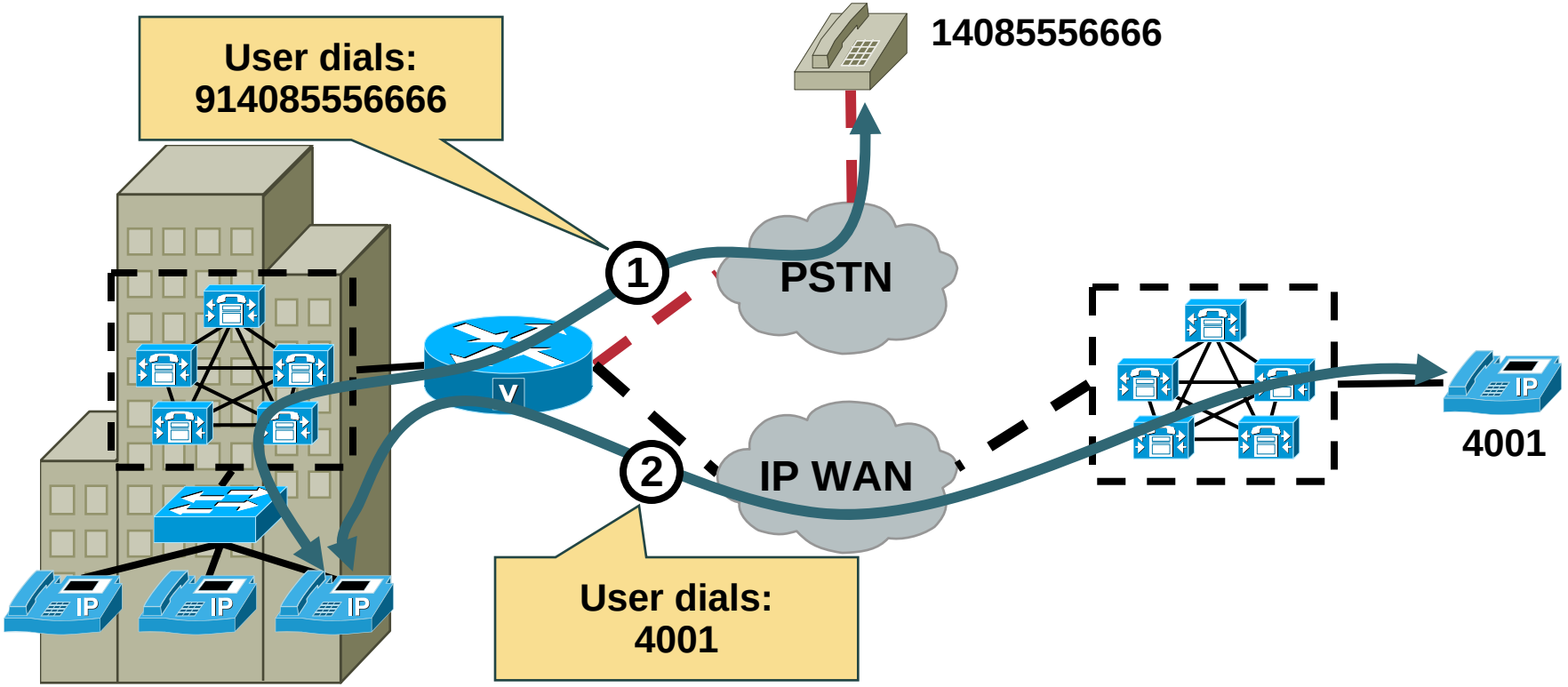
Route Patterns



	Route Pattern	Partition	Destination	Digit Manipulation
1	9.!#	SJC-PSTN	Router1	Discard 9
2	4XXX	IntersiteRouting	Atlanta CM Trunk	None

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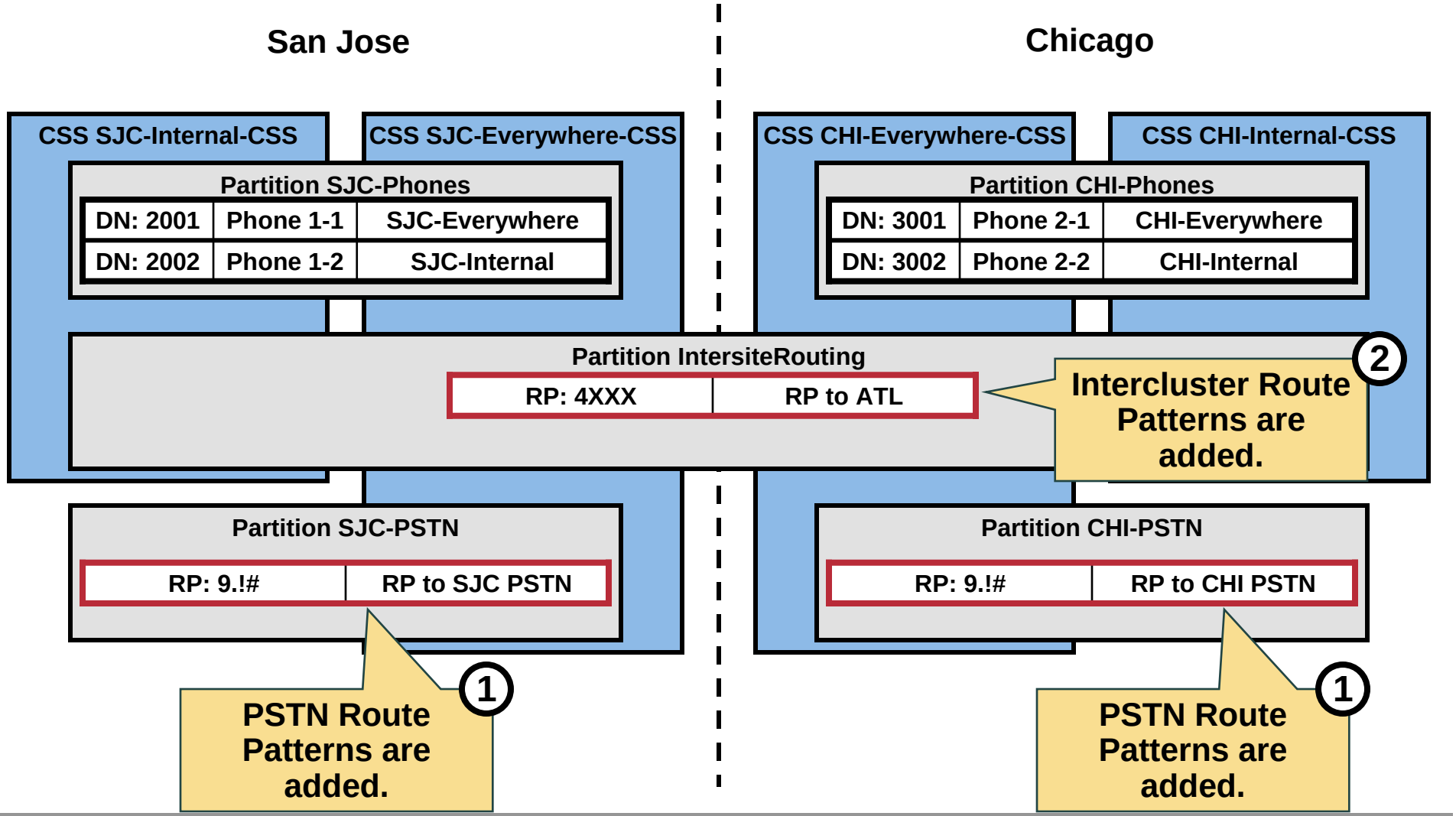
Route Patterns



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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!**

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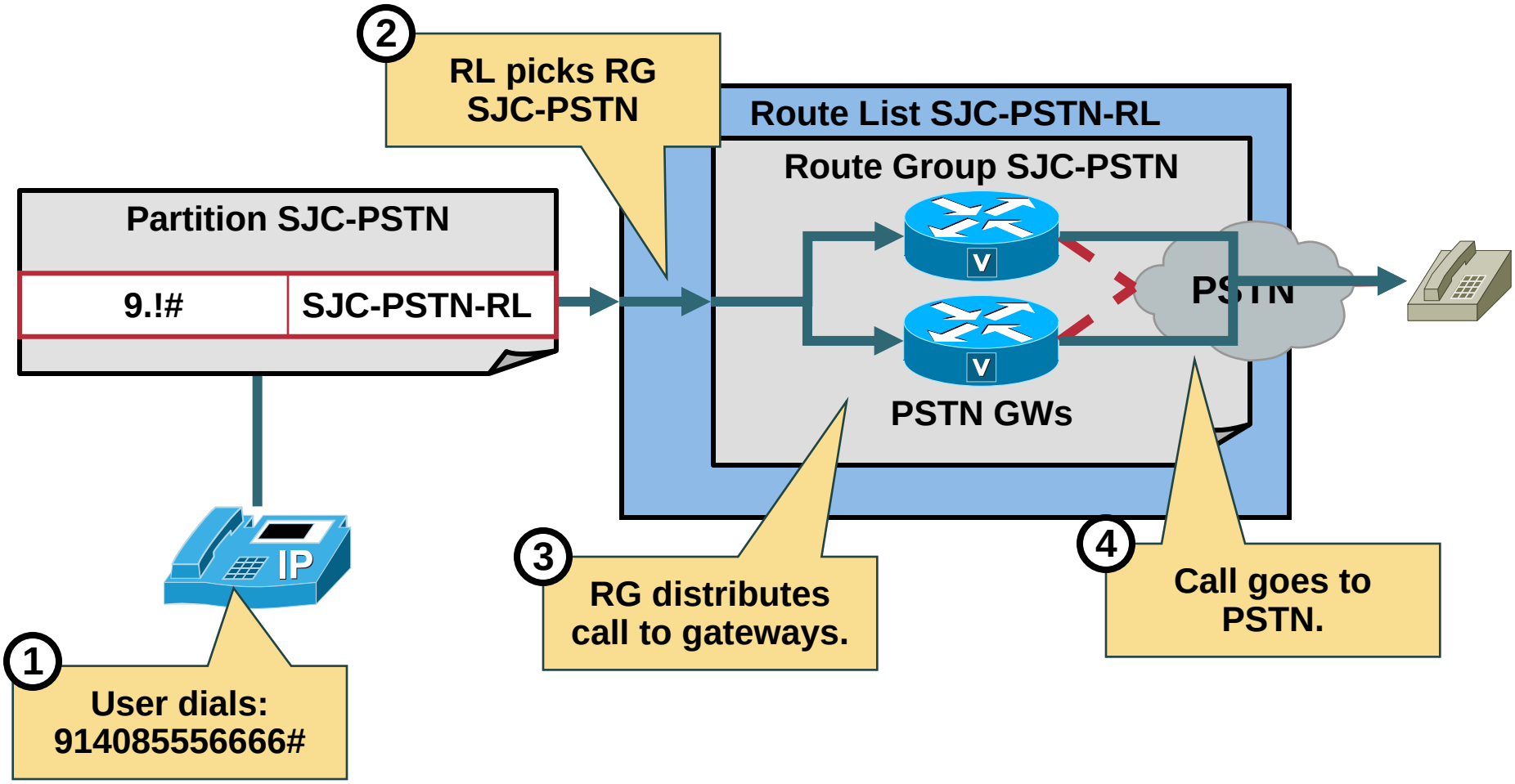
Route Lists and Route Groups



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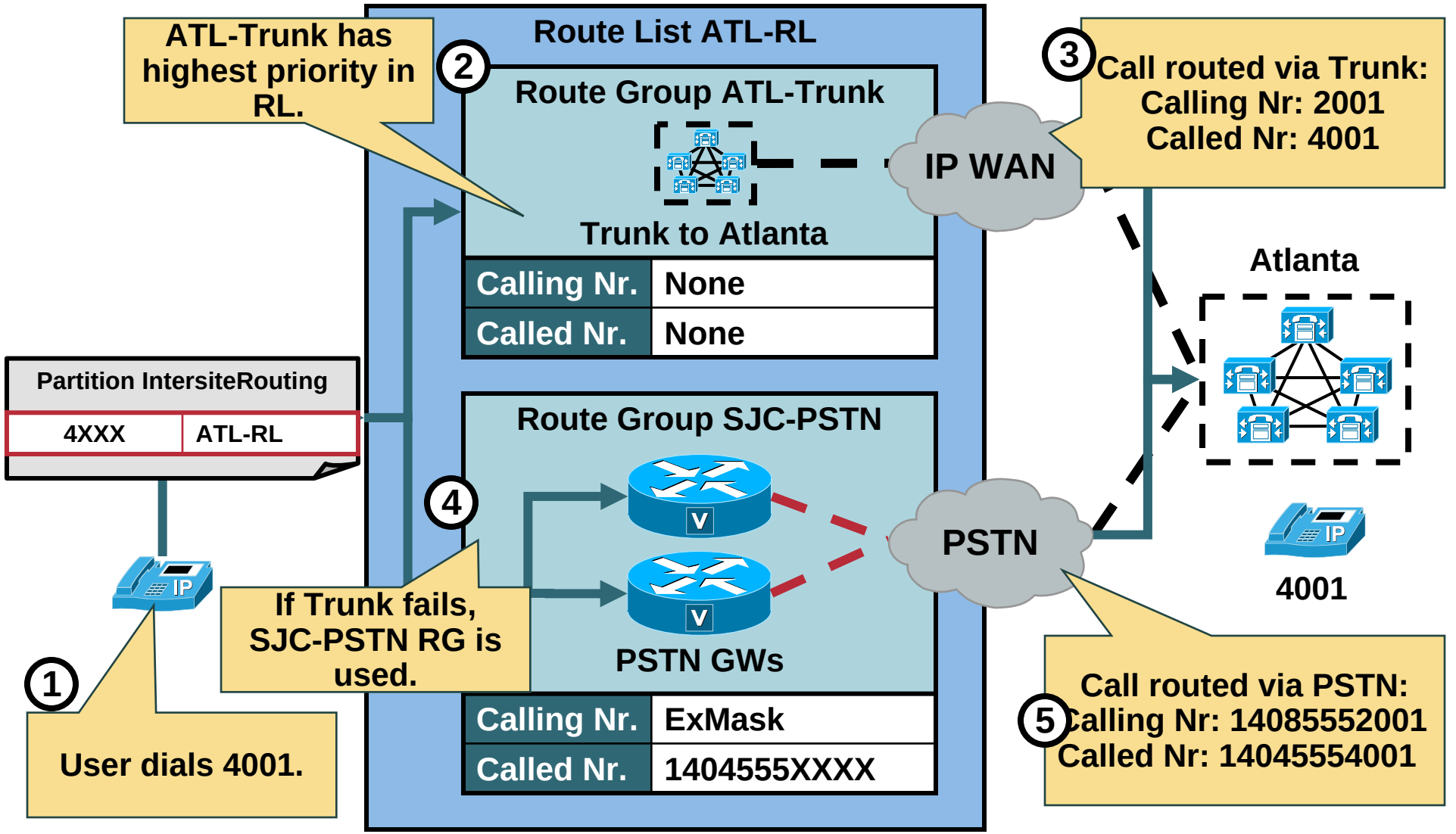
Route Lists and Route Groups



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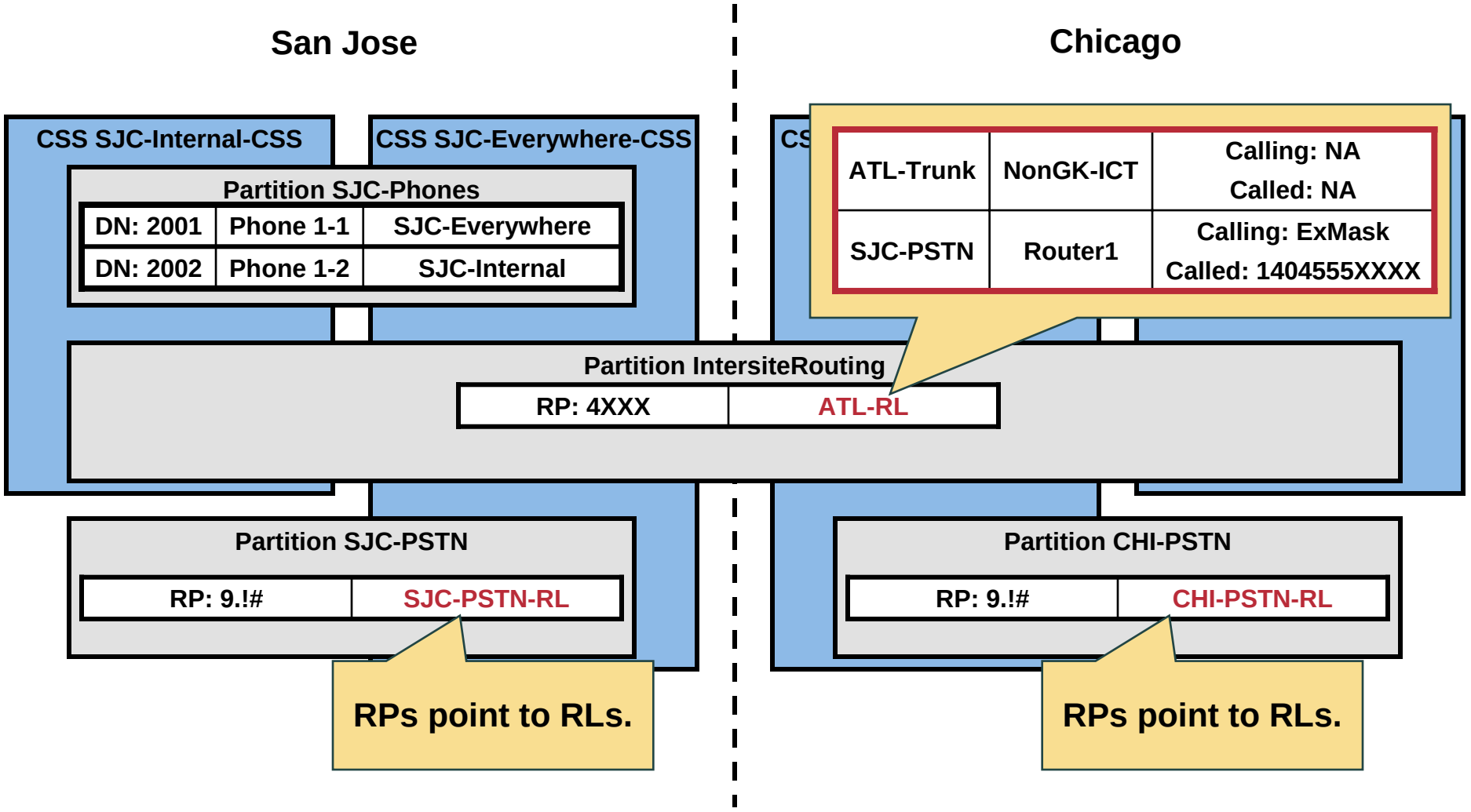
Route Lists and Route Groups



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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.**

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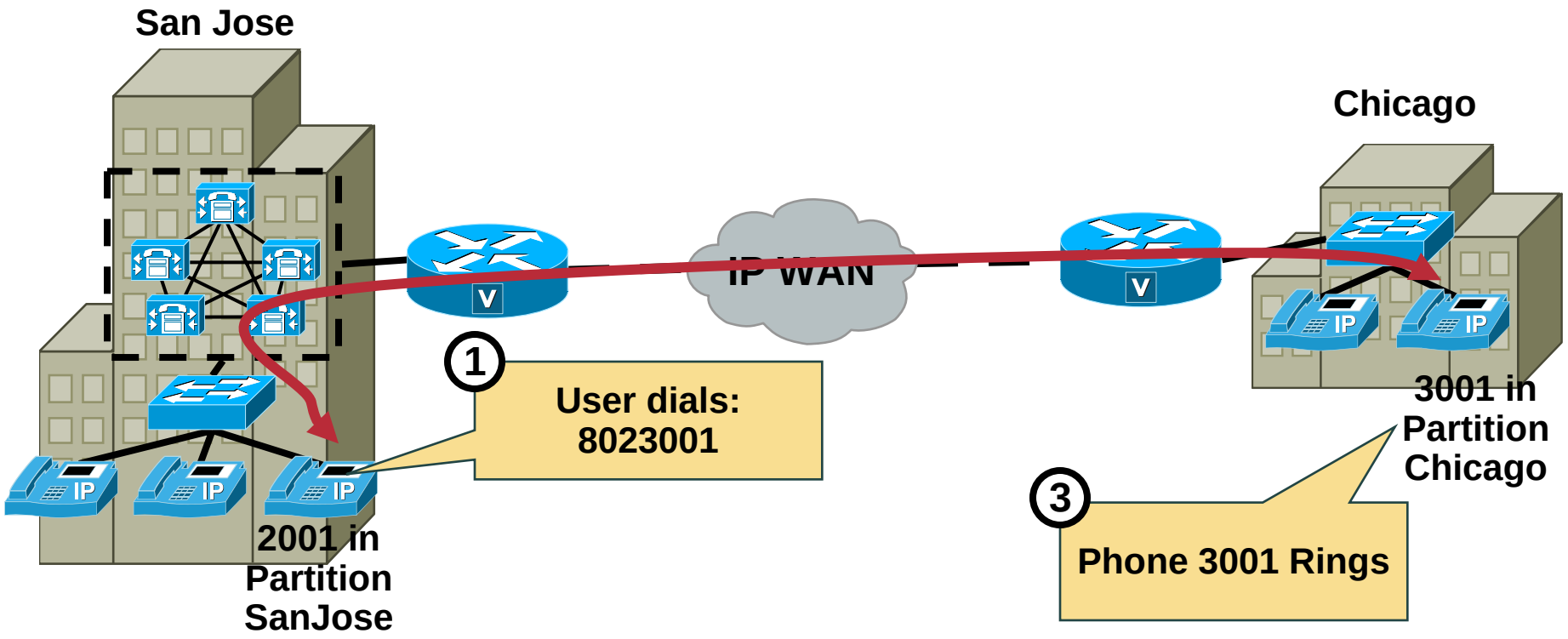
Translation Patterns



2 Match found: 8023XXX

Translation Pattern	Partition	Destination	Digit Manipulation
8012XXX	IntersiteRouting	SJC-Phones-CSS	Discard 801
8023XXX	IntersiteRouting	CHI-Phones-CSS	Discard 802

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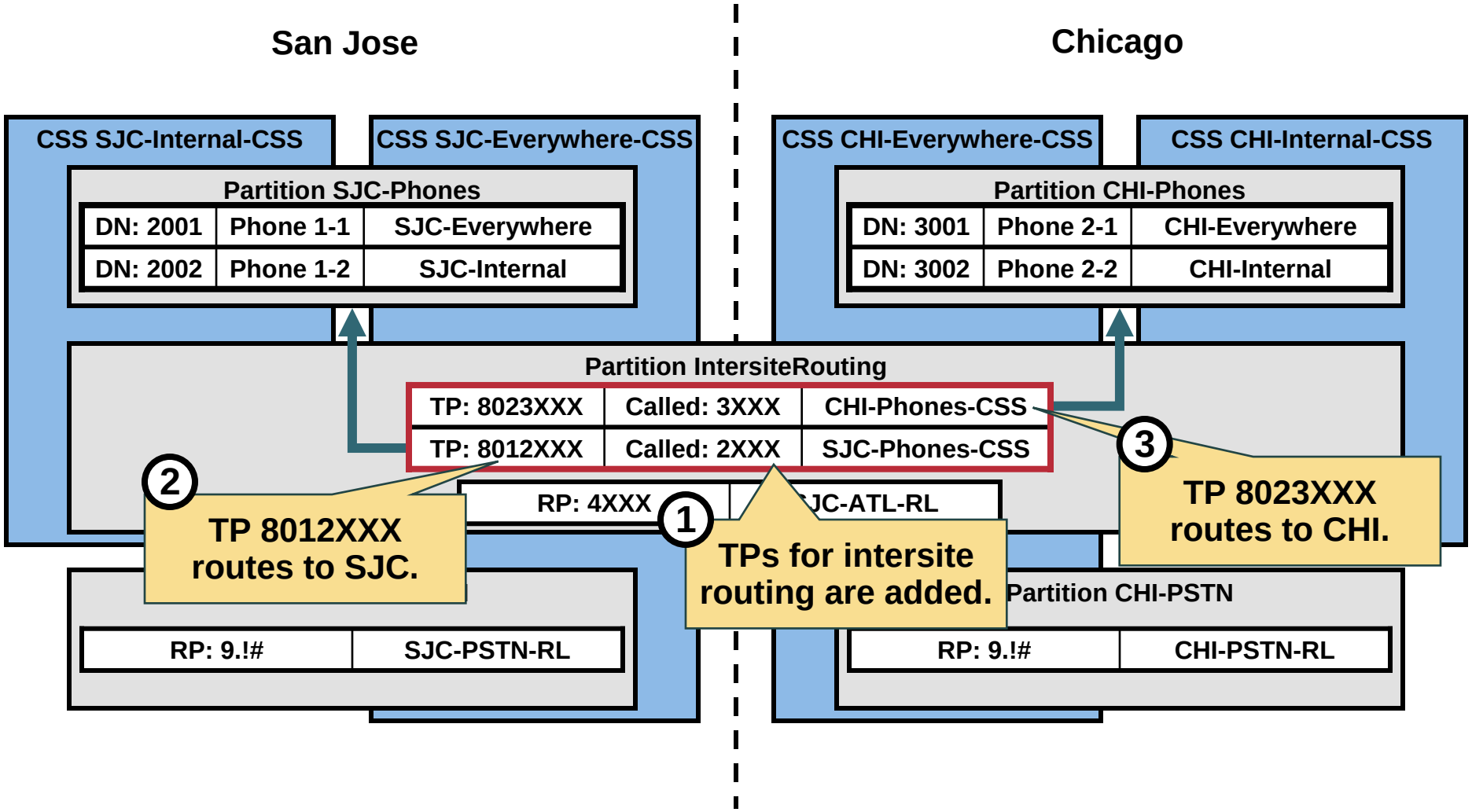
Translation Patterns



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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.**

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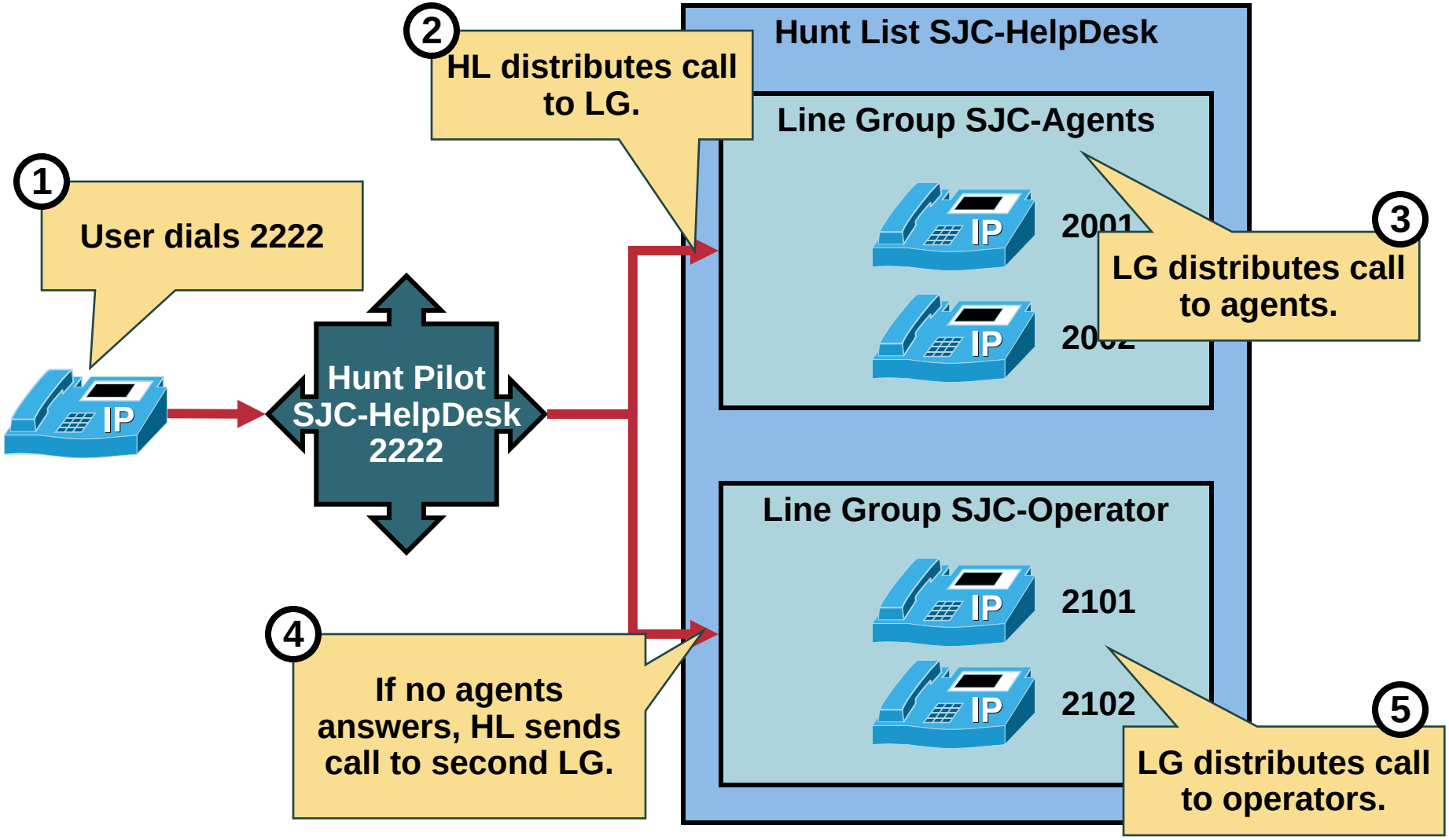
Line Groups, Hunt Lists and Hunt Pilots



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Review of Cisco Unified CallManager Dial Plan Components

T. E.1

Dial Plan Functionality	CCM Component
Endpoint Addressing	•Directory Numbers •Partitions
Call Routing and Path Selection	•Partitions and Calling Search Spaces •Route Patterns •Route Groups and Route Lists •Translation Patterns
Digit Manipulation	•Route Patterns •Route Groups and Route Lists •Translation Patterns
Call Privileges	•Partitions •Calling Search Spaces
Call Coverage	•Line Group, Hunt List and Hunt Pilot

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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 diferentiates 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.

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Deploying a Dial Plan on CallManager 5.0

Configuring CallManager Voice Gateways

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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

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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.**

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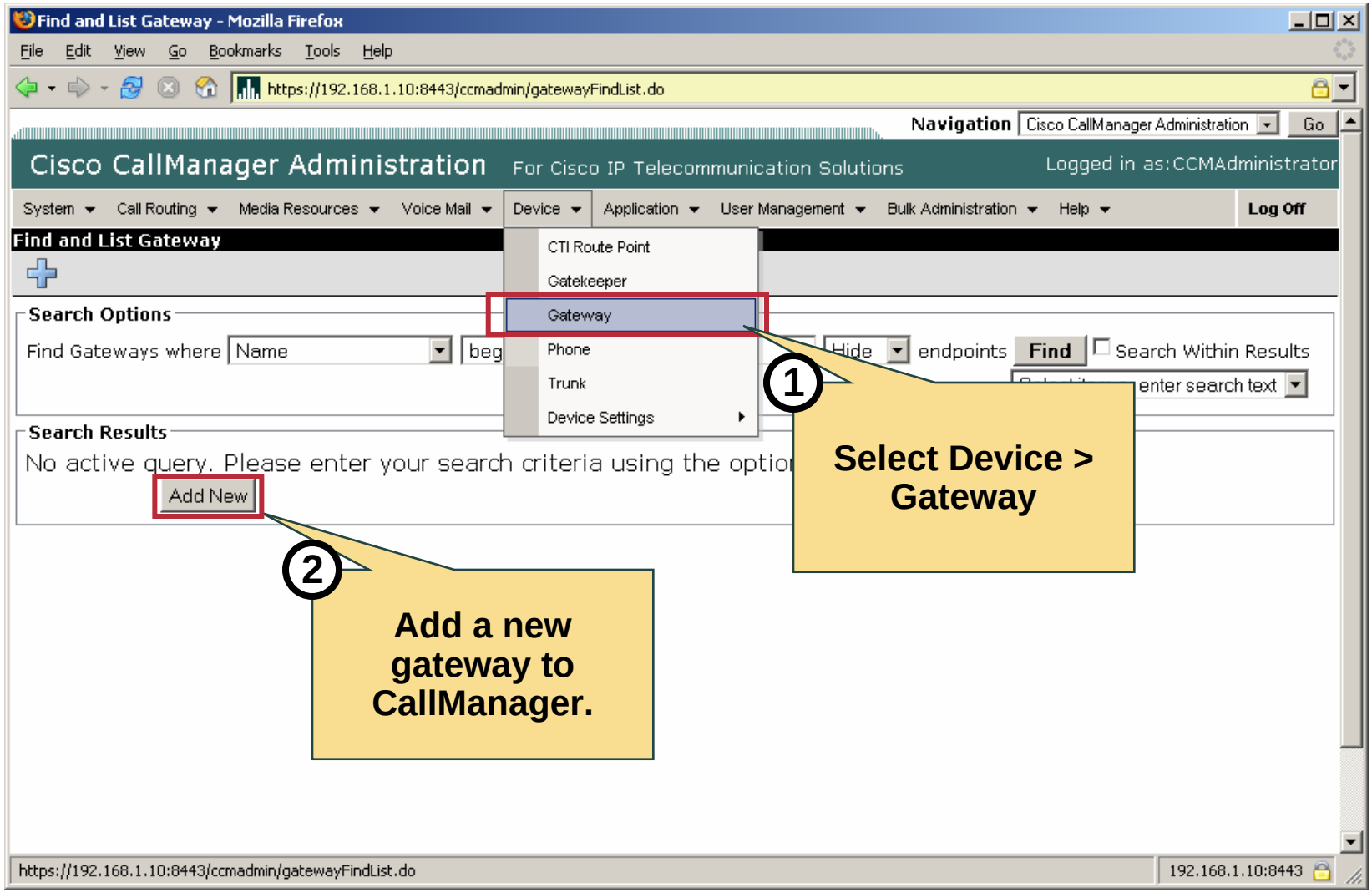
Configuring MGCP Gateways in Cisco Unified CallManager: Add Gateway



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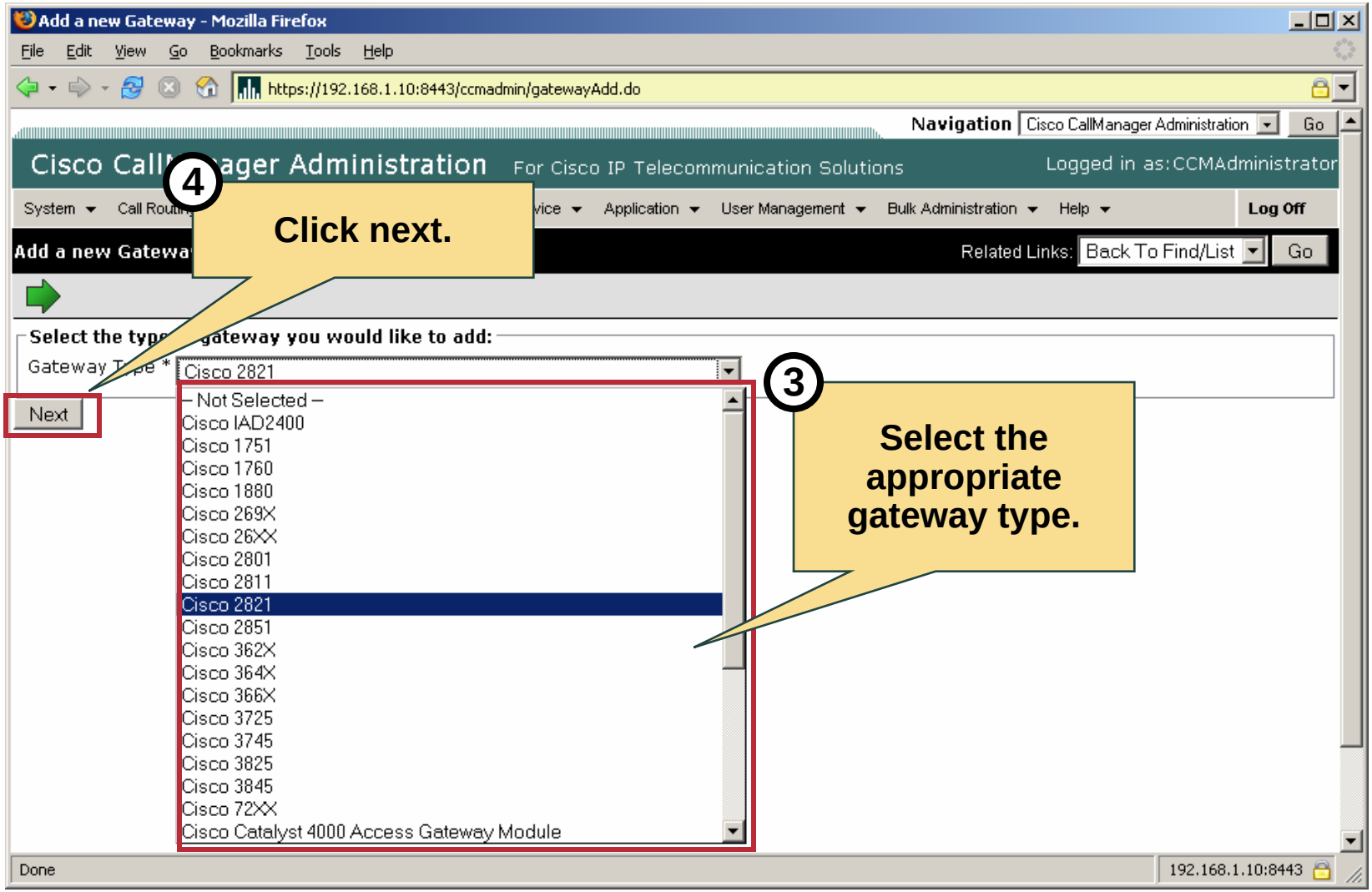
Configuring MGCP Gateways in Cisco Unified CallManager: Add Gateway



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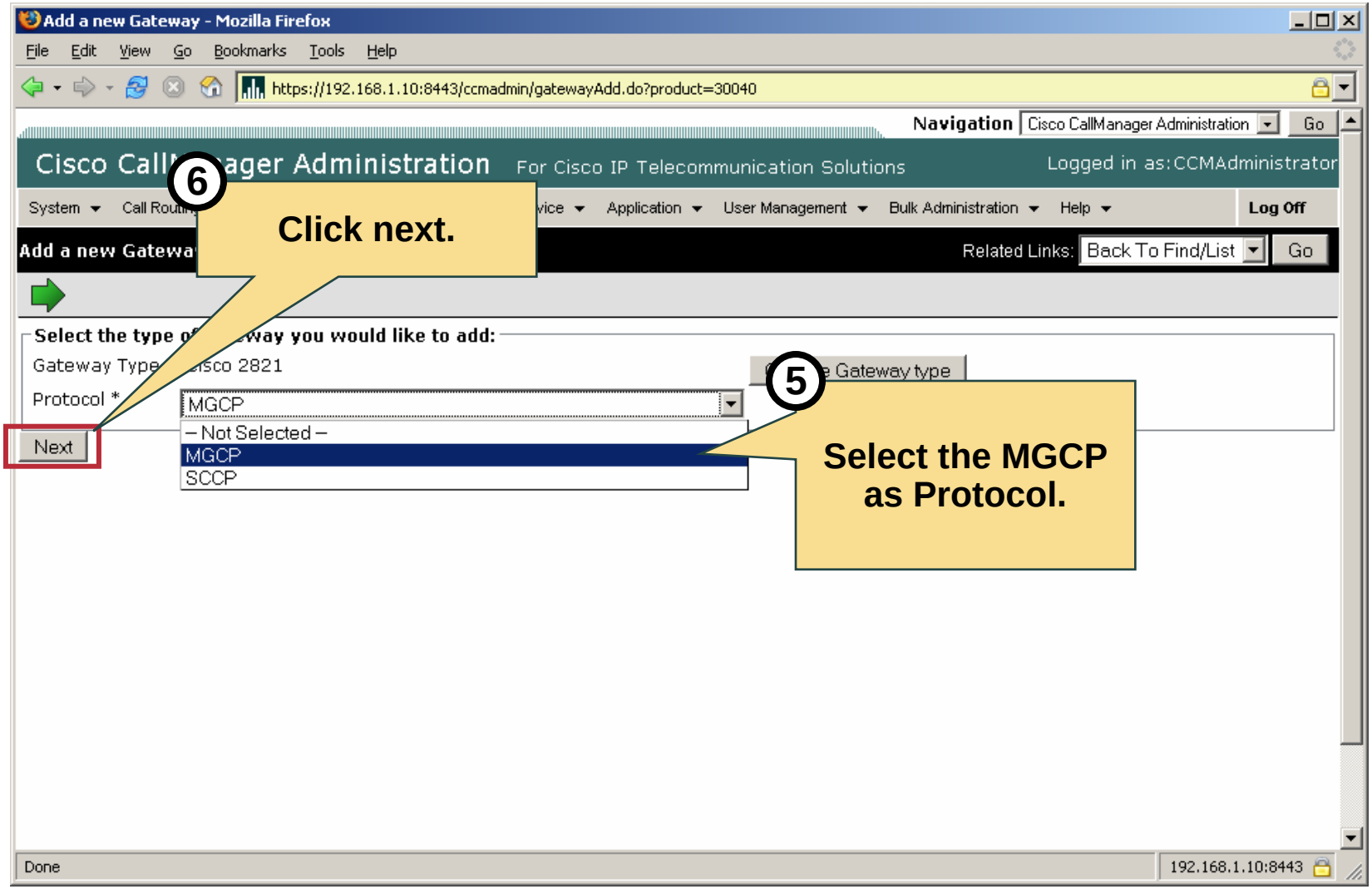
Configuring MGCP Gateways in Cisco Unified CallManager: Add Gateway



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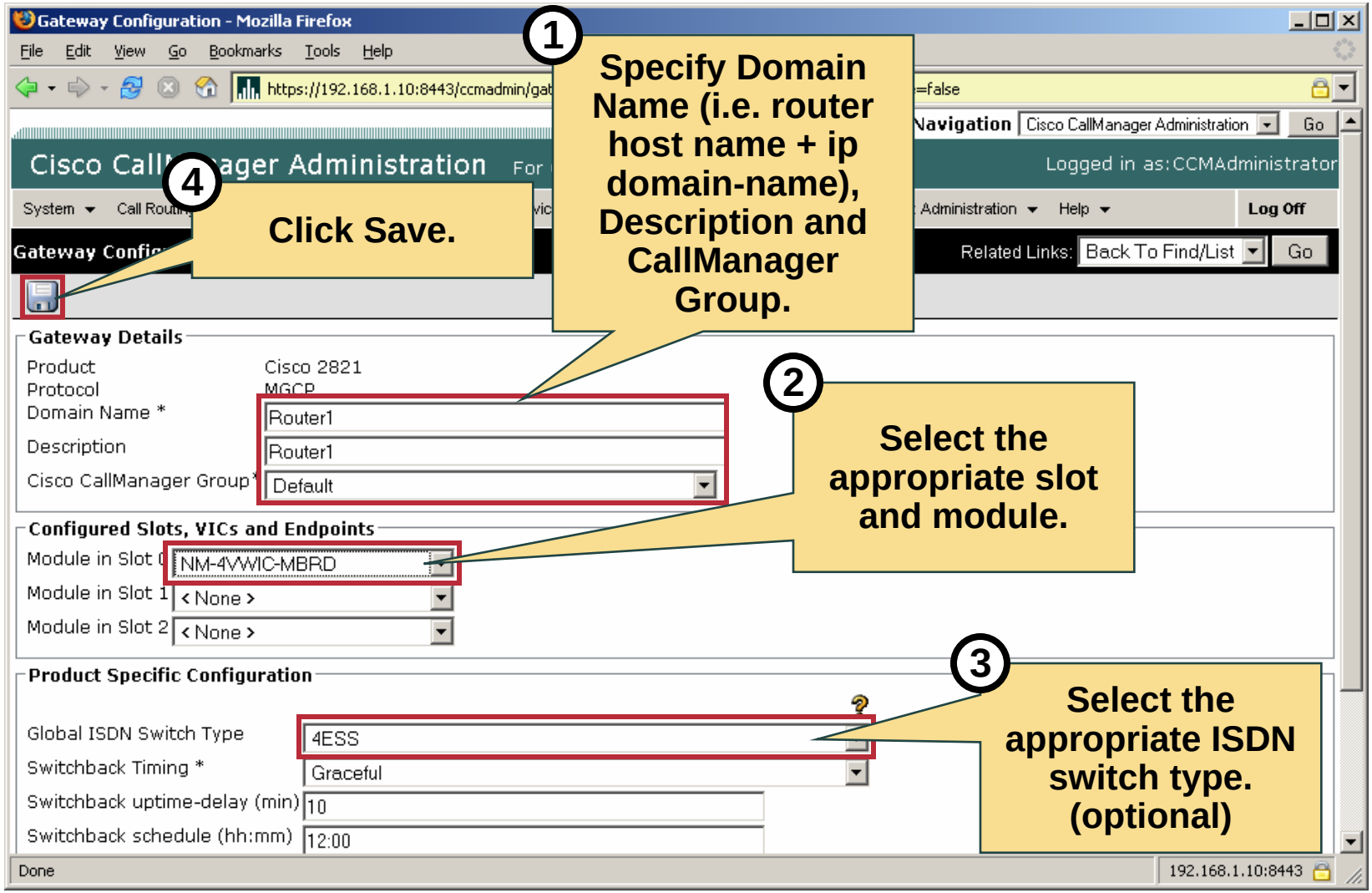
Configuring MGCP Gateways in Cisco Unified CallManager: Add Slot



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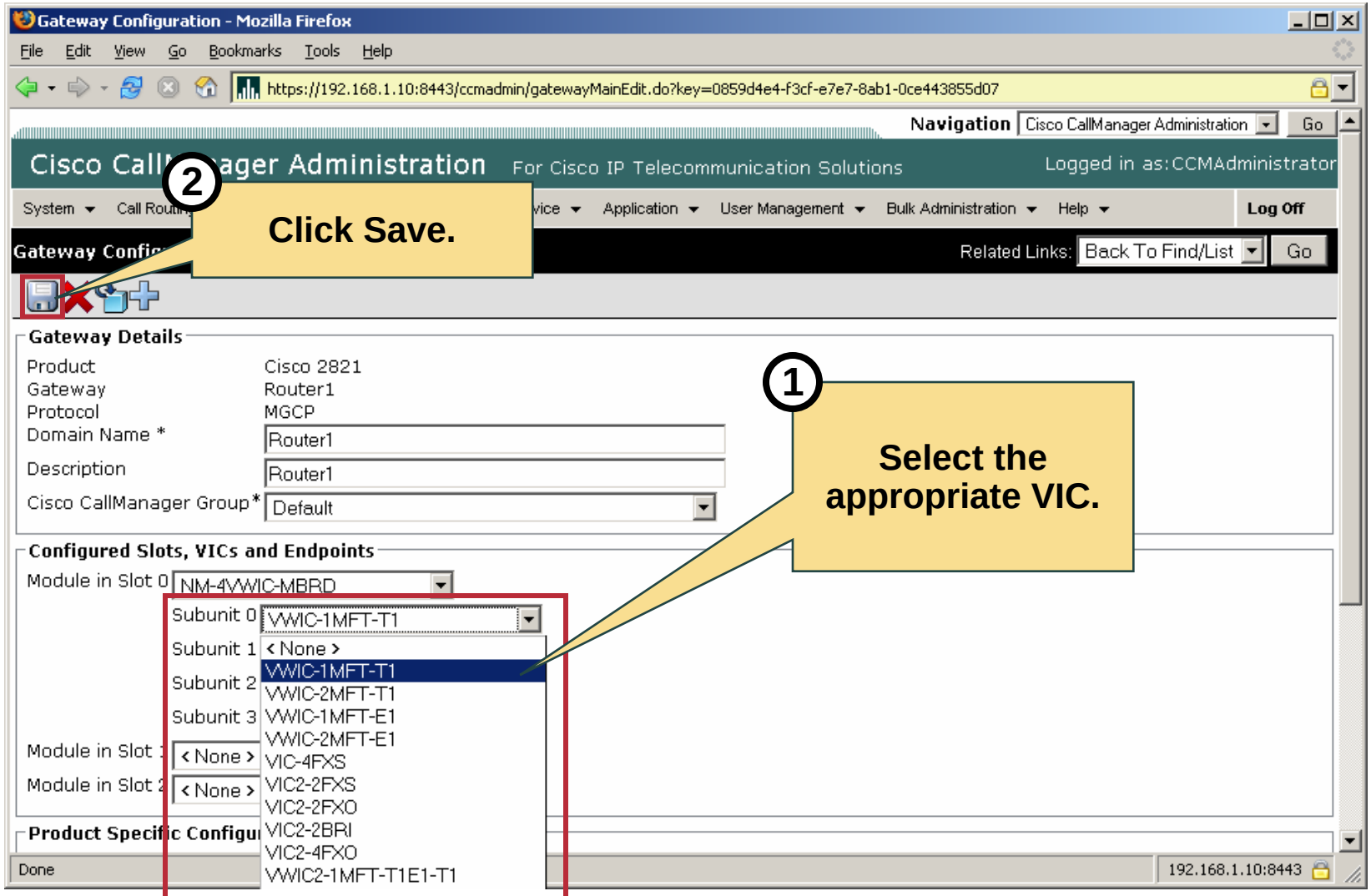
Configuring MGCP Gateways in Cisco Unified CallManager: Add VICs



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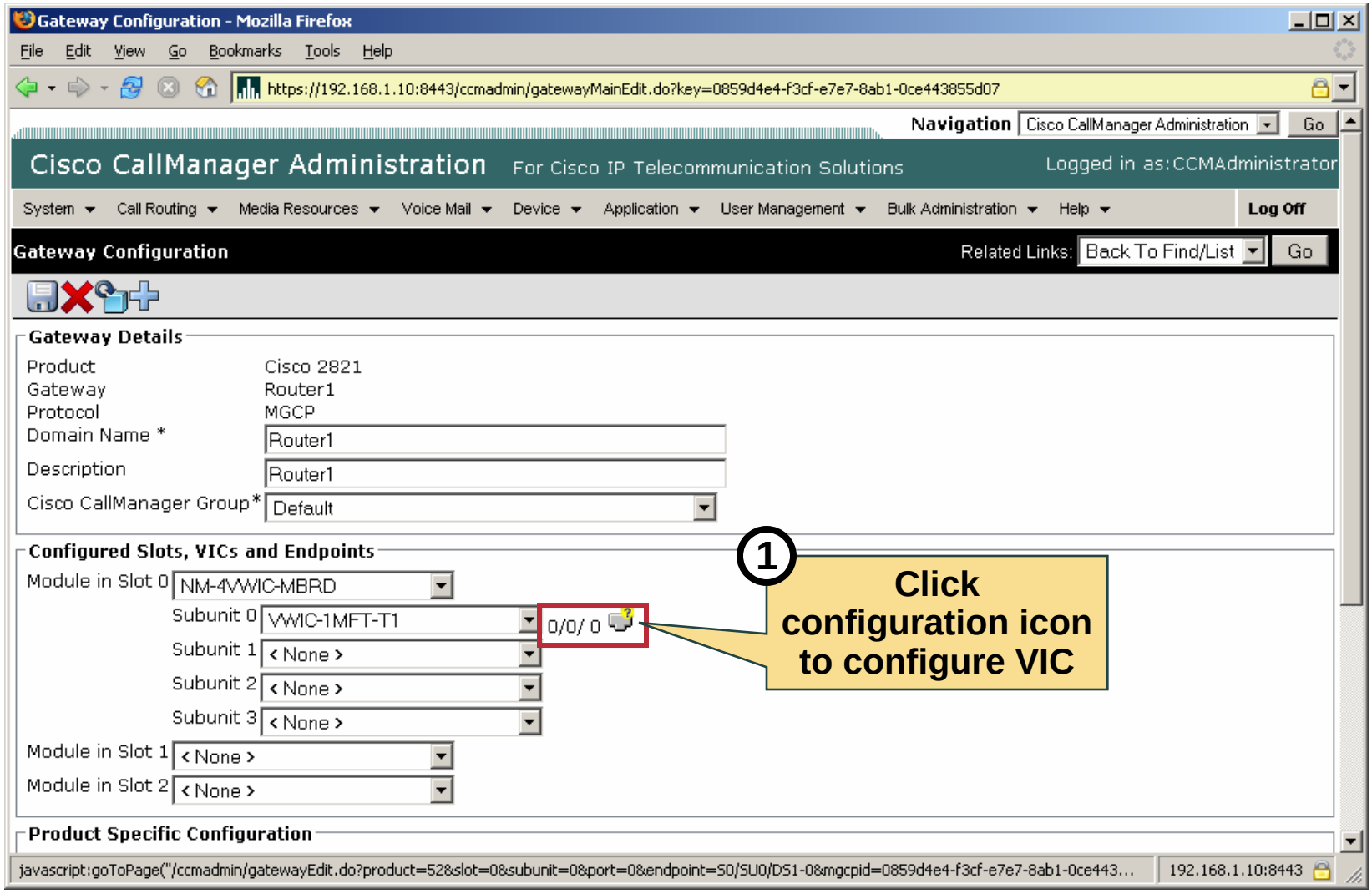
Configuring MGCP Gateways in Cisco Unified CallManager: Configure VICs



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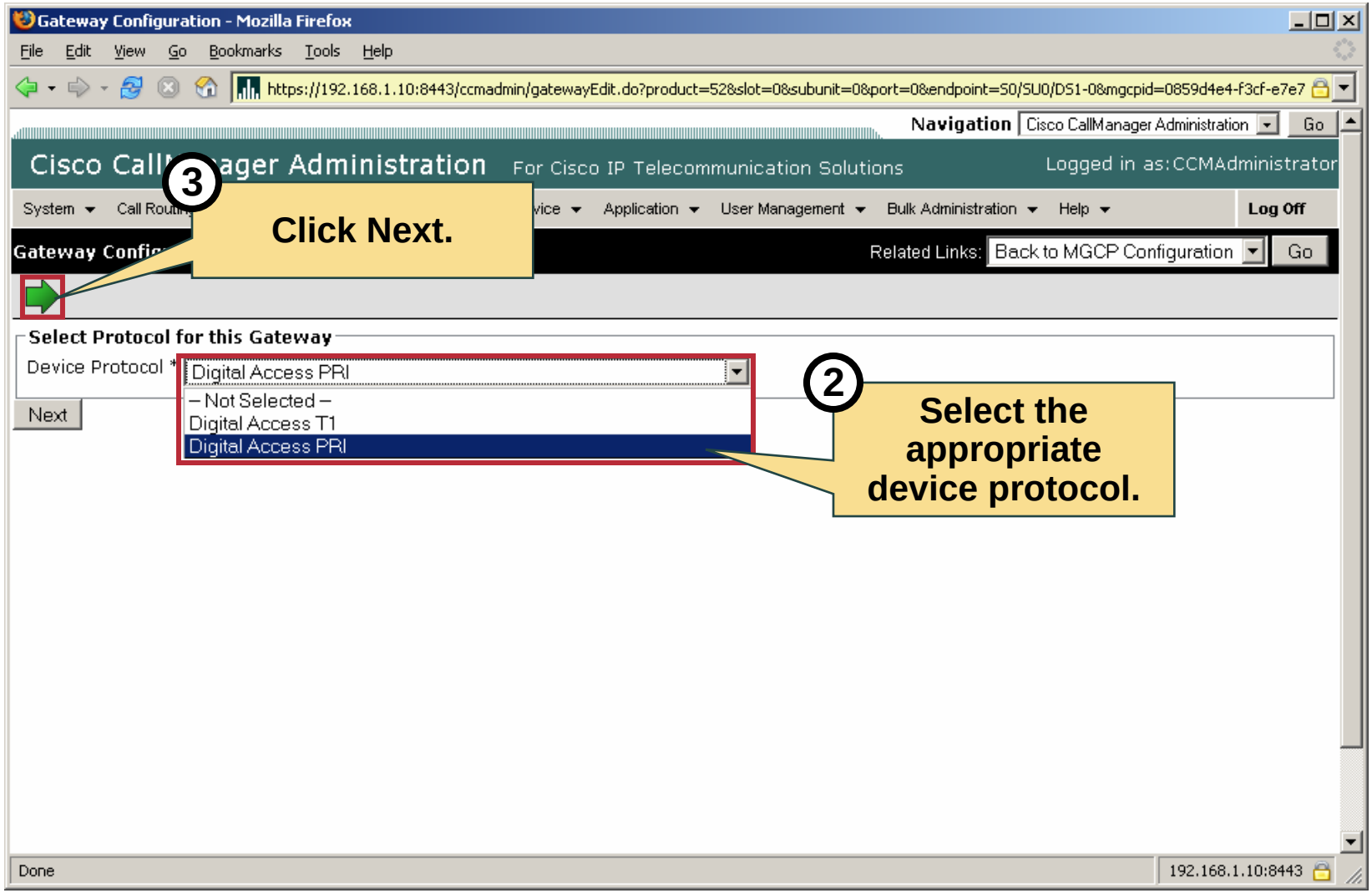
Configuring MGCP Gateways in Cisco Unified CallManager: Configure VICs



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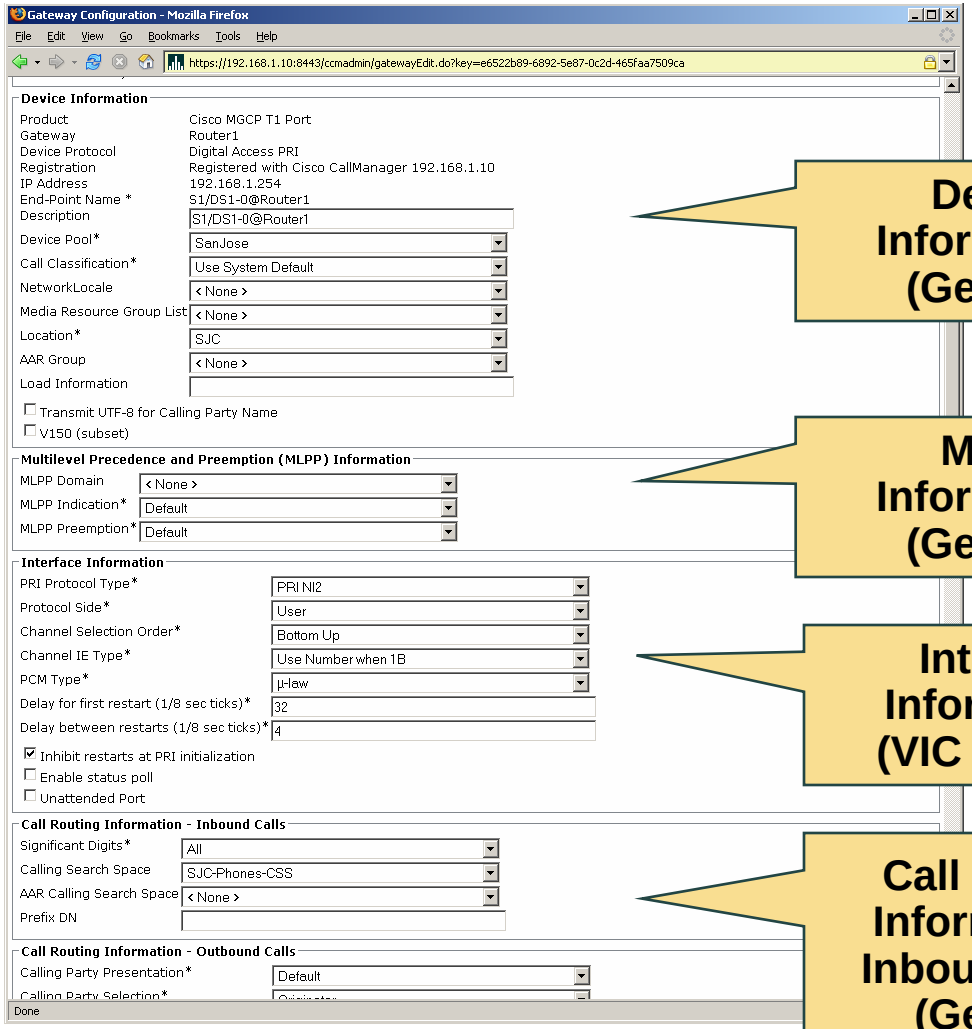
Configuring MGCP Gateways in Cisco Unified CallManager: Configure VICs



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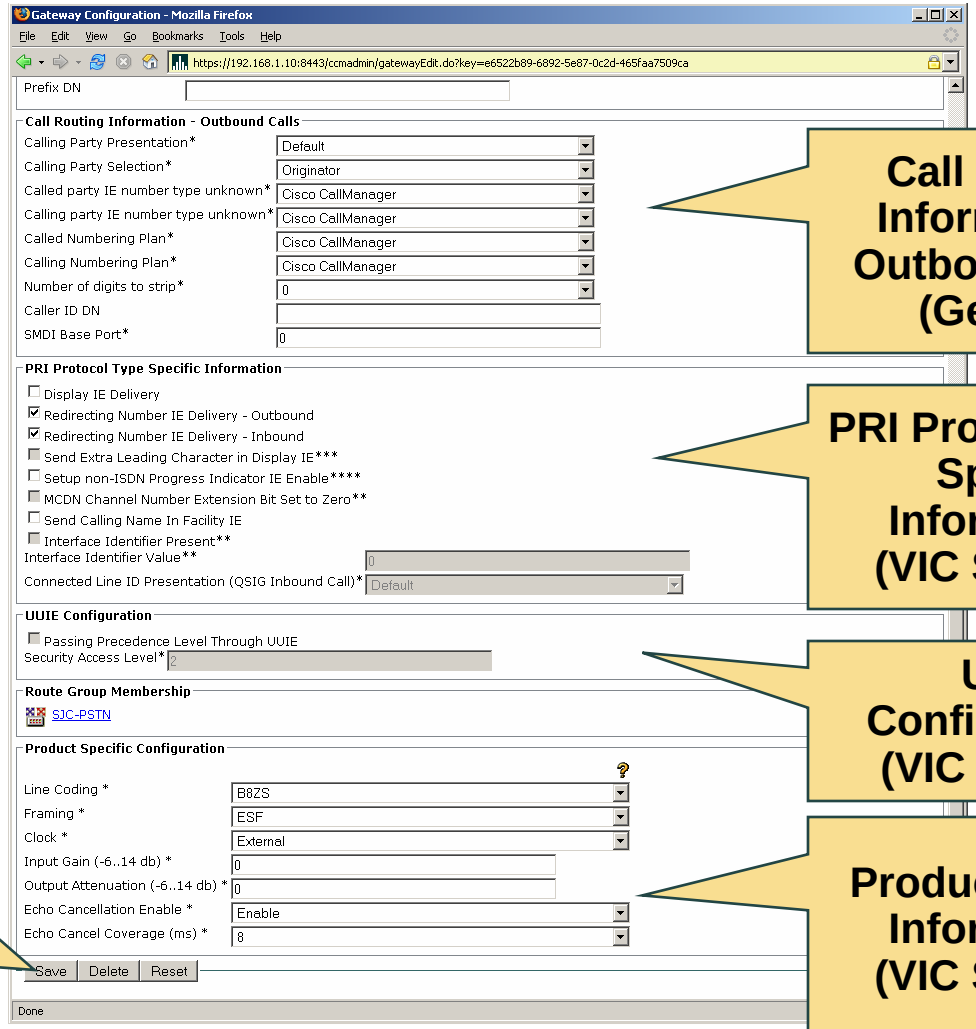
Configuring MGCP Gateways in Cisco Unified CallManager: Configure VICs



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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

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Configuring Gateways for MGCP

IOS MGCP gateway configuration steps:

1. Specify configuration server.
2. Enable configuration download.

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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**

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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.

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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

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 with CallManager.

Registered MGCP endpoints.

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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:  0x807FFFFF
    Number of L2 Discards = 0, L2 Session ID = 2
    Total Allocated ISDN CCBs = 0
…
```

Backhauled to CallManager.

Layer 2 up.

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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.

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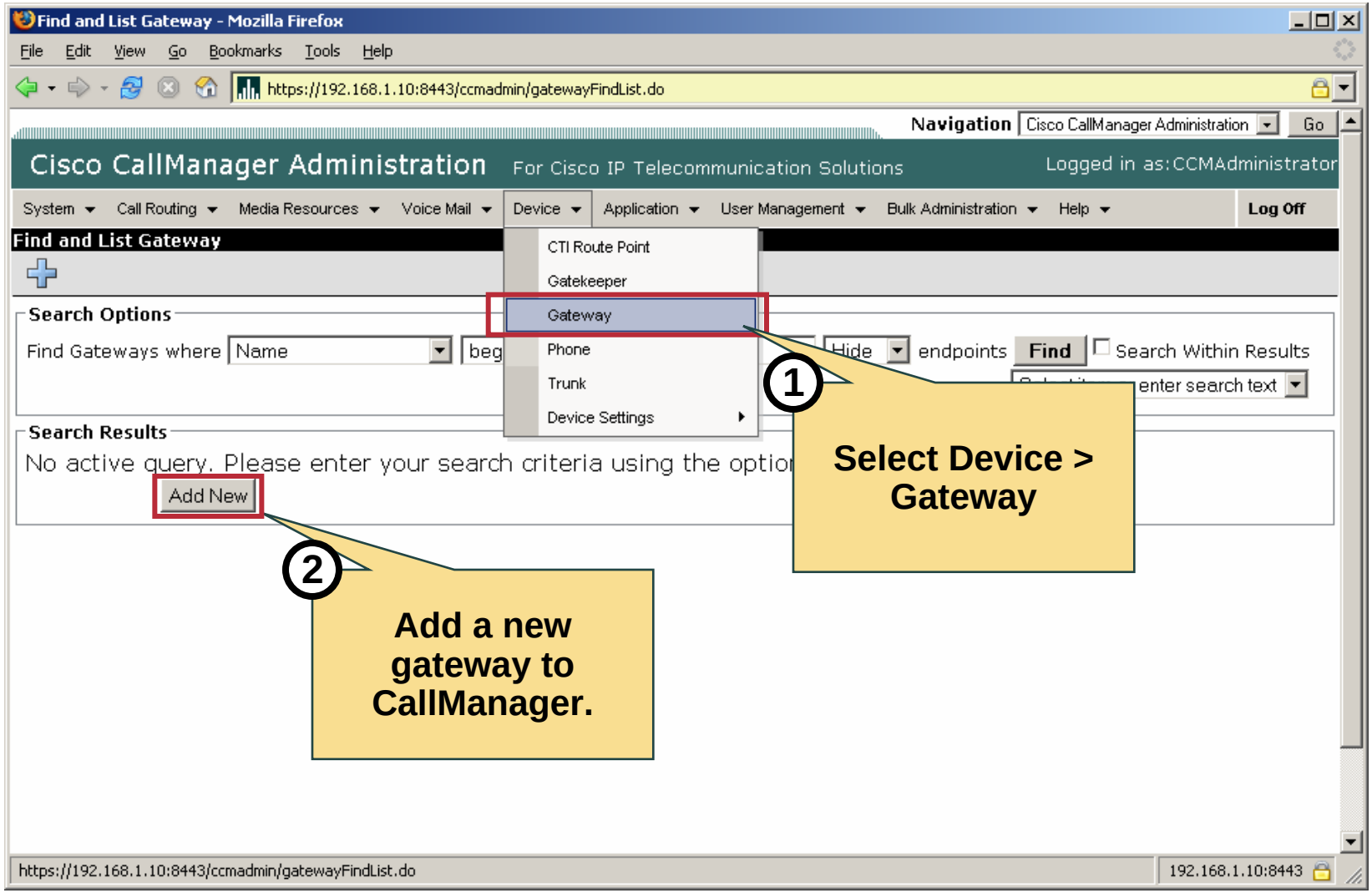
Configuring H.323 Gateways in Cisco Unified CallManager: Add Gateway



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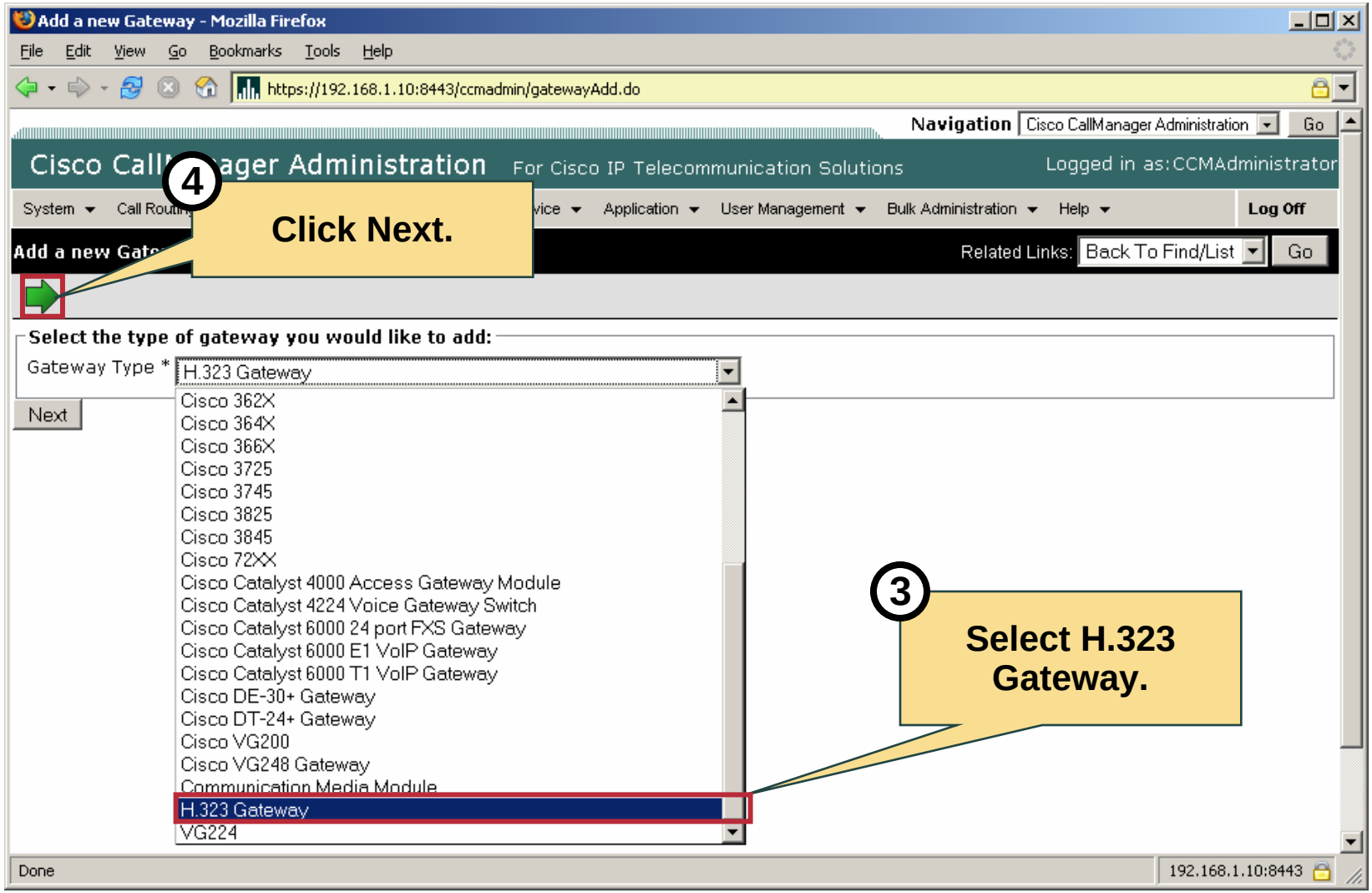
Configuring H.323 Gateways in Cisco Unified CallManager: Add Gateway



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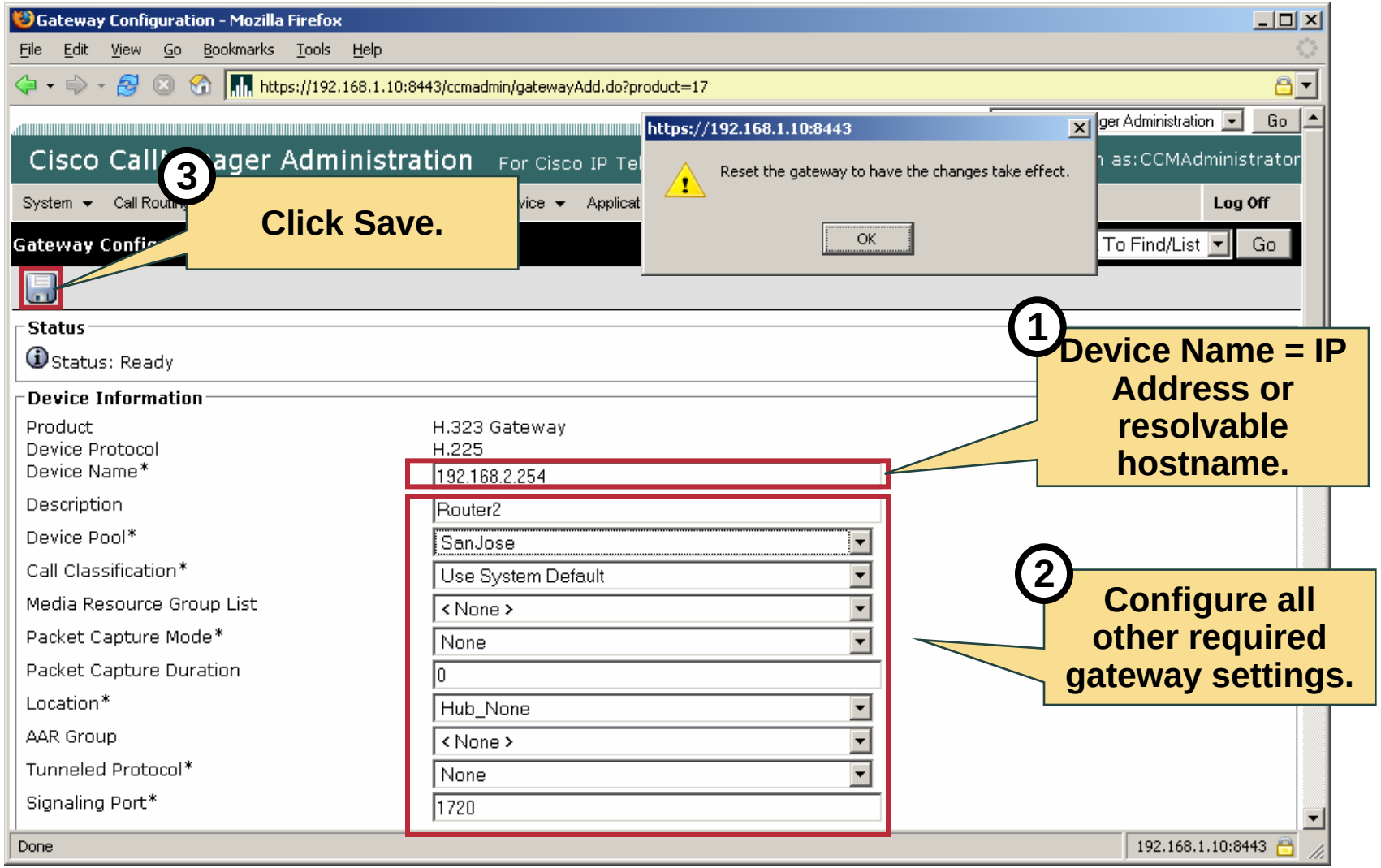
Configuring H.323 Gateways in Cisco Unified CallManager: Gateway



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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

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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.

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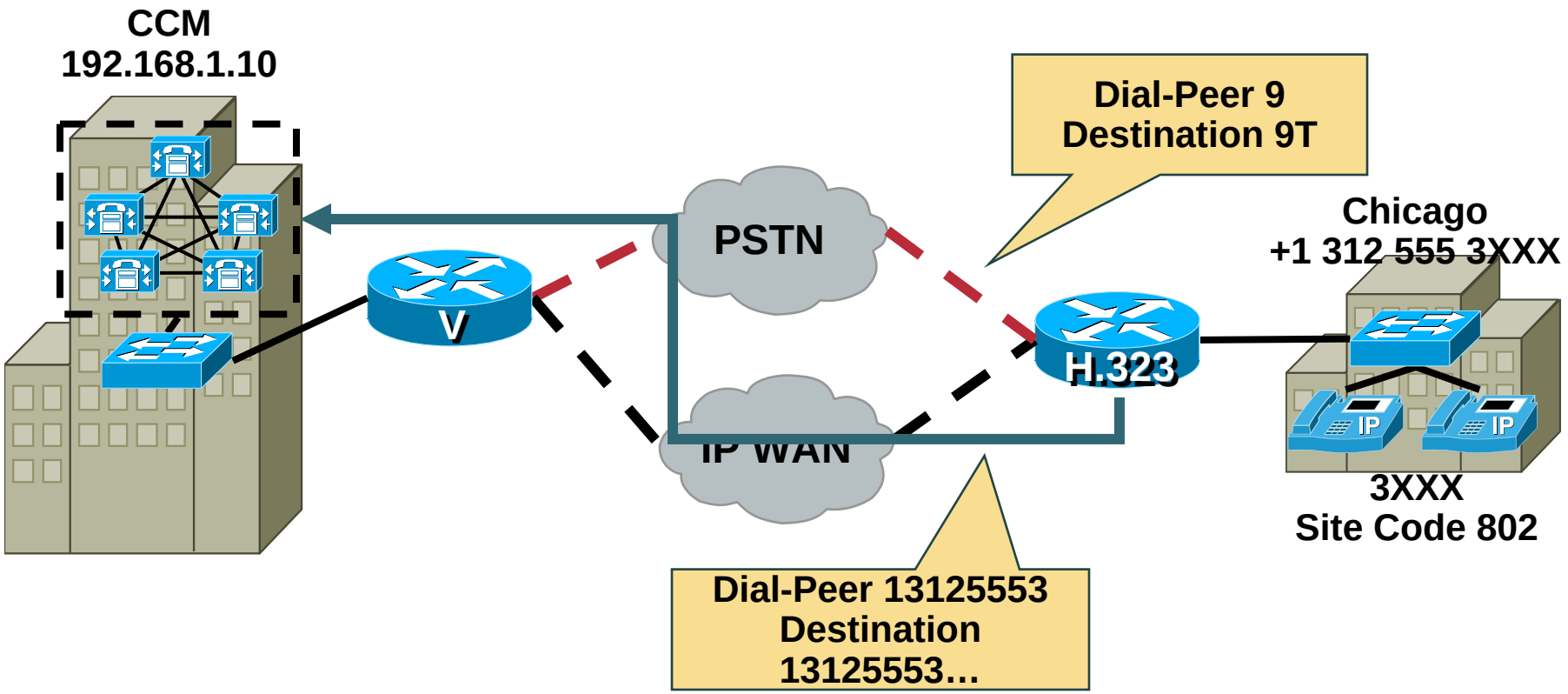
Configuring Gateways for H.323



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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
!
```

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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
!
```

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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

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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.

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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
!
```

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Verifying H.323 Gateway Operation

```
router#
show voice port summary
```

- **Displays voice-port information.**

```
router#
debug isdn q931
```

- **Displays ISDN Q931 information.**

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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
…
```

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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
      }
    }
…
```

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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.

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Deploying a Dial Plan on CallManager 5.0

Configuring Cisco Unified CallManager Trunks

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Objectives

- Trunk Overview
- Configuring Non-Gatekeeper Controlled Intercluster Trunks
- Configuring Gatekeeper Controlled Intercluster Trunks
- Configuring QSIG for Trunks
- Configuring SIP Trunks

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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

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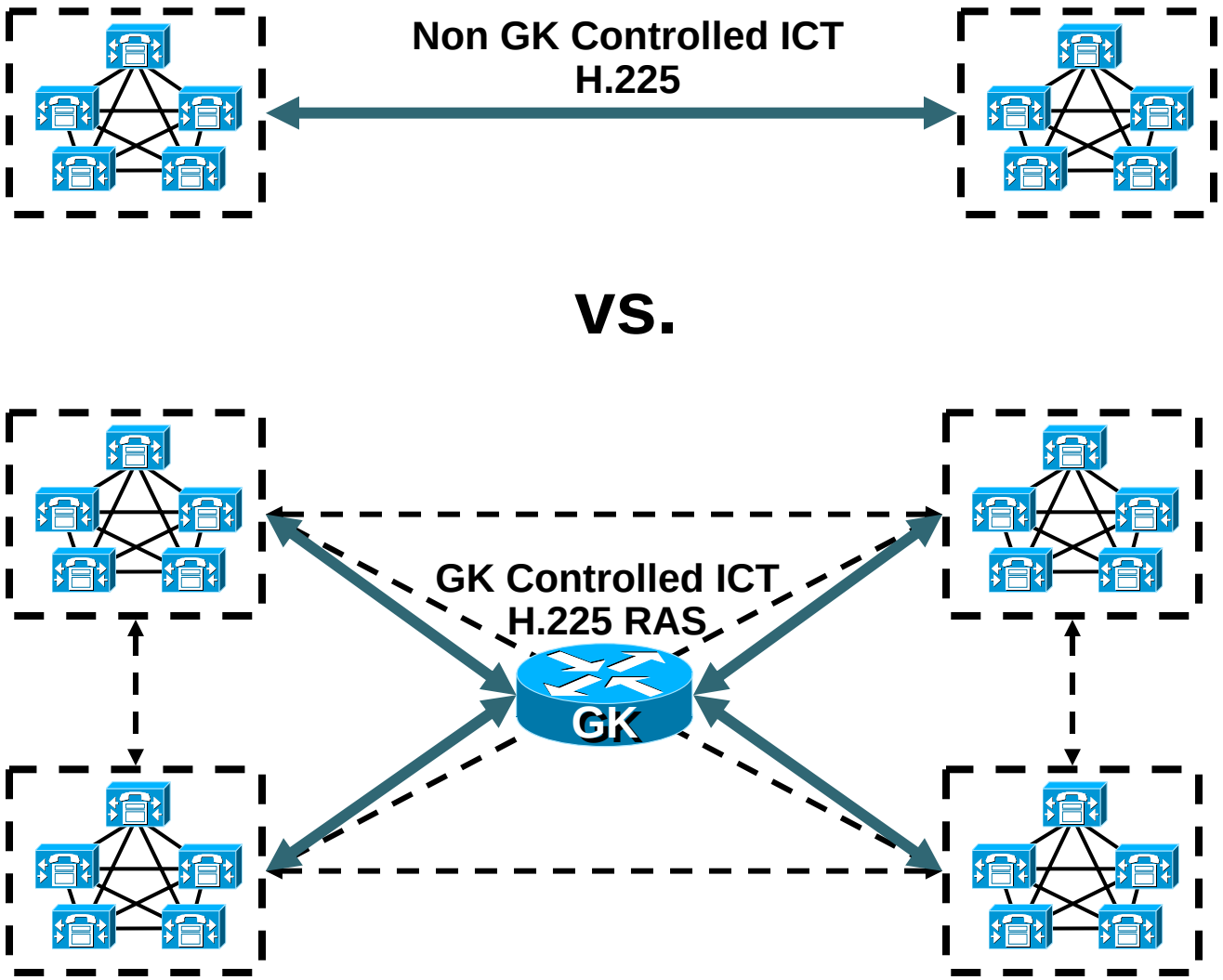
Trunk Overview



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Configuring Non-Gatekeeper Controlled Intercluster Trunks

Non-Gatekeeper Controlled Intercluster trunk configuration steps:

1. **Add Non-Gatekeeper Controlled intercluster trunk.**
2. **Specify trunk parameters.**

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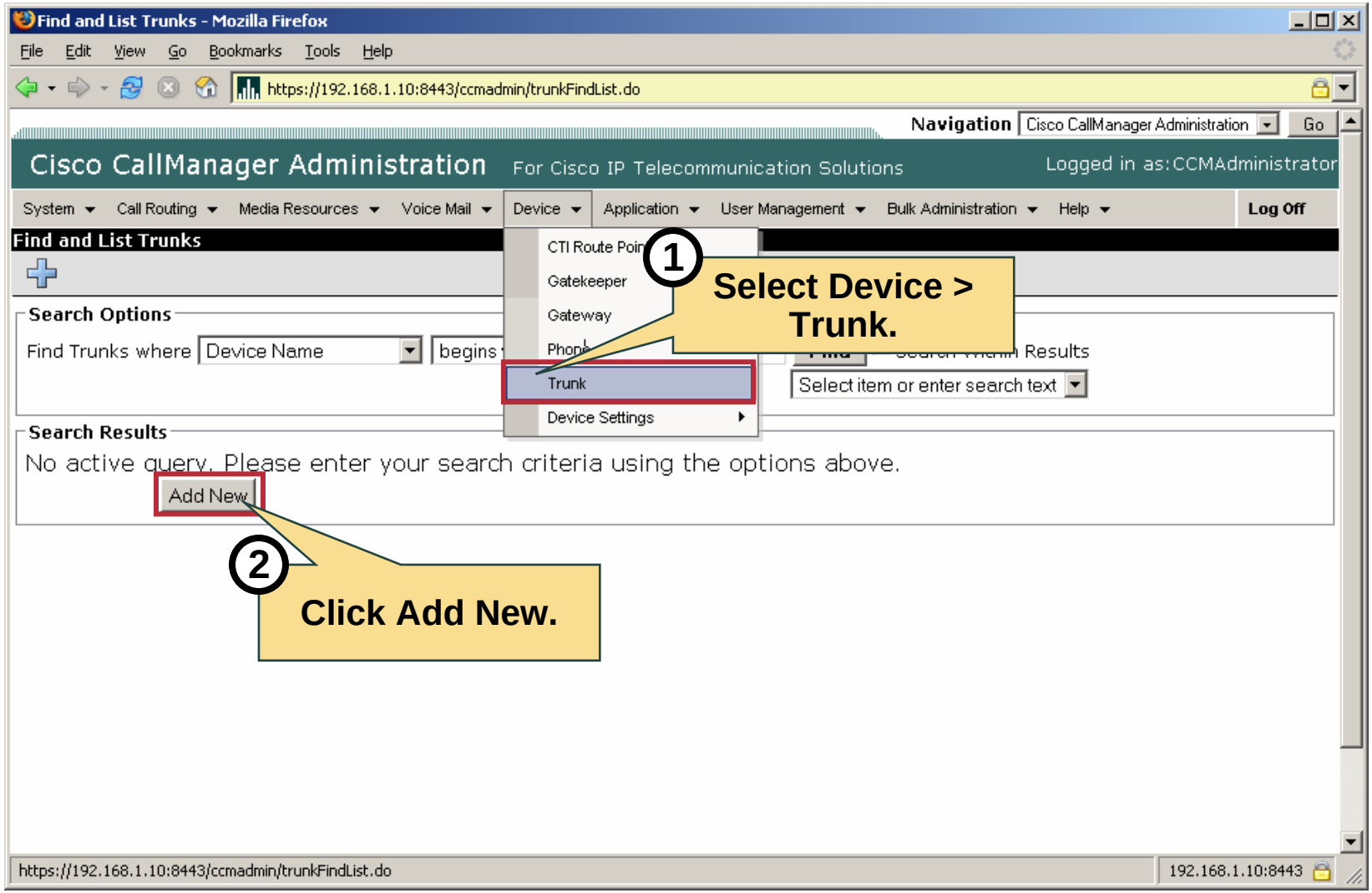
Configuring Non-Gatekeeper Controlled Intercluster Trunks: Adding a Trunk



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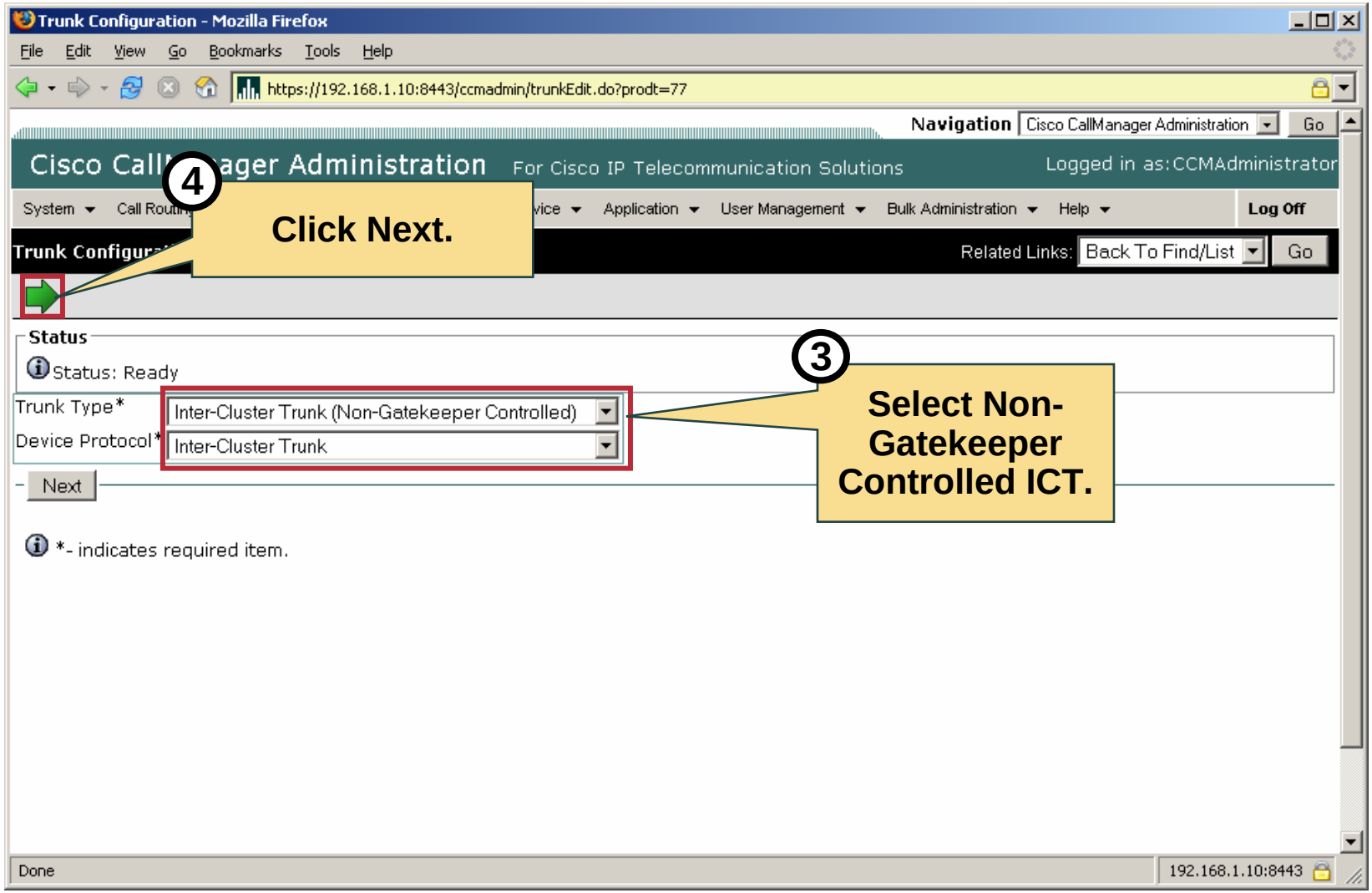
Configuring Non-Gatekeeper Controlled Intercluster Trunks: Adding a Trunk



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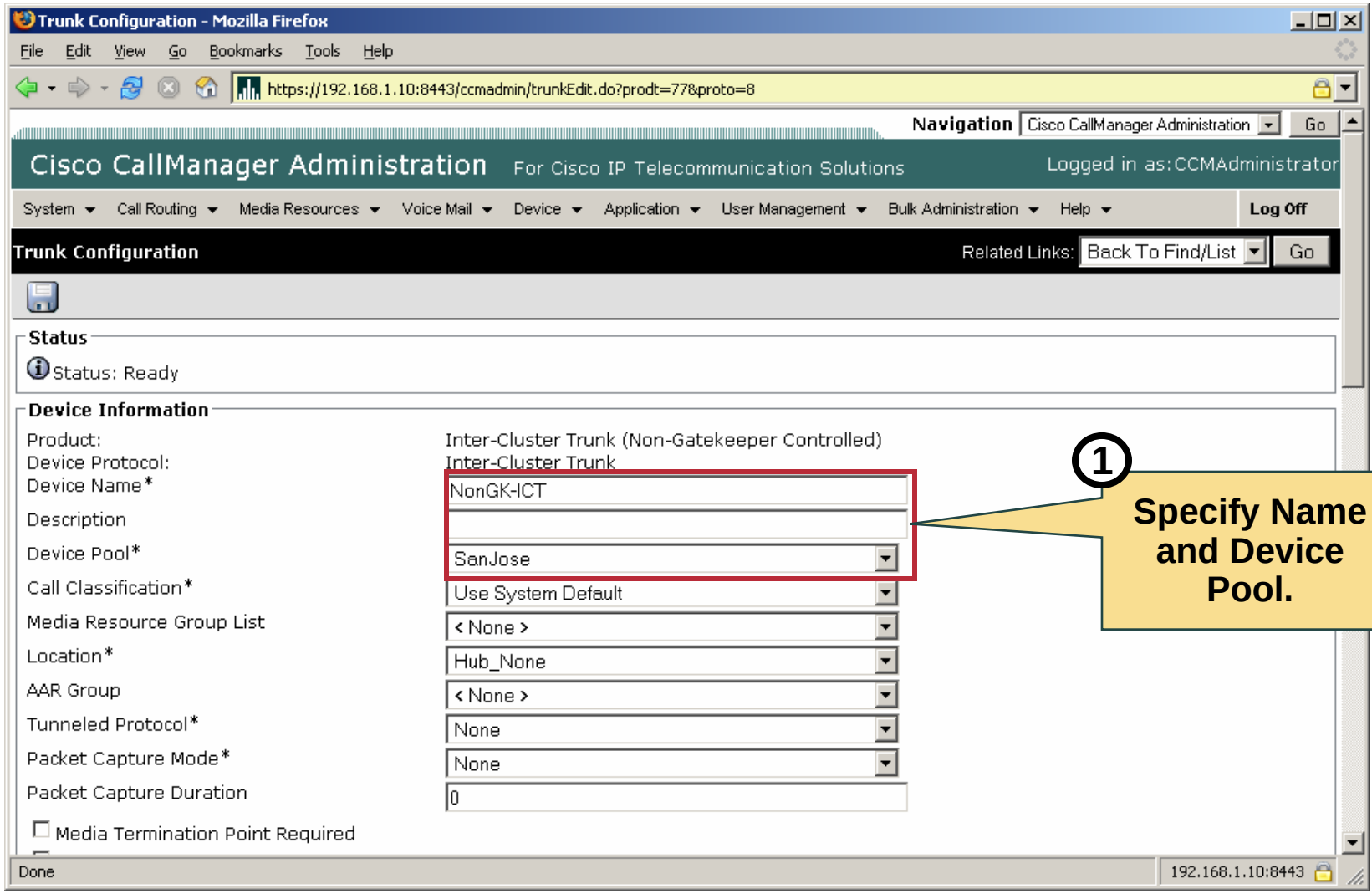
Configuring Non-Gatekeeper Controlled Intercluster Trunks: Configuring Options



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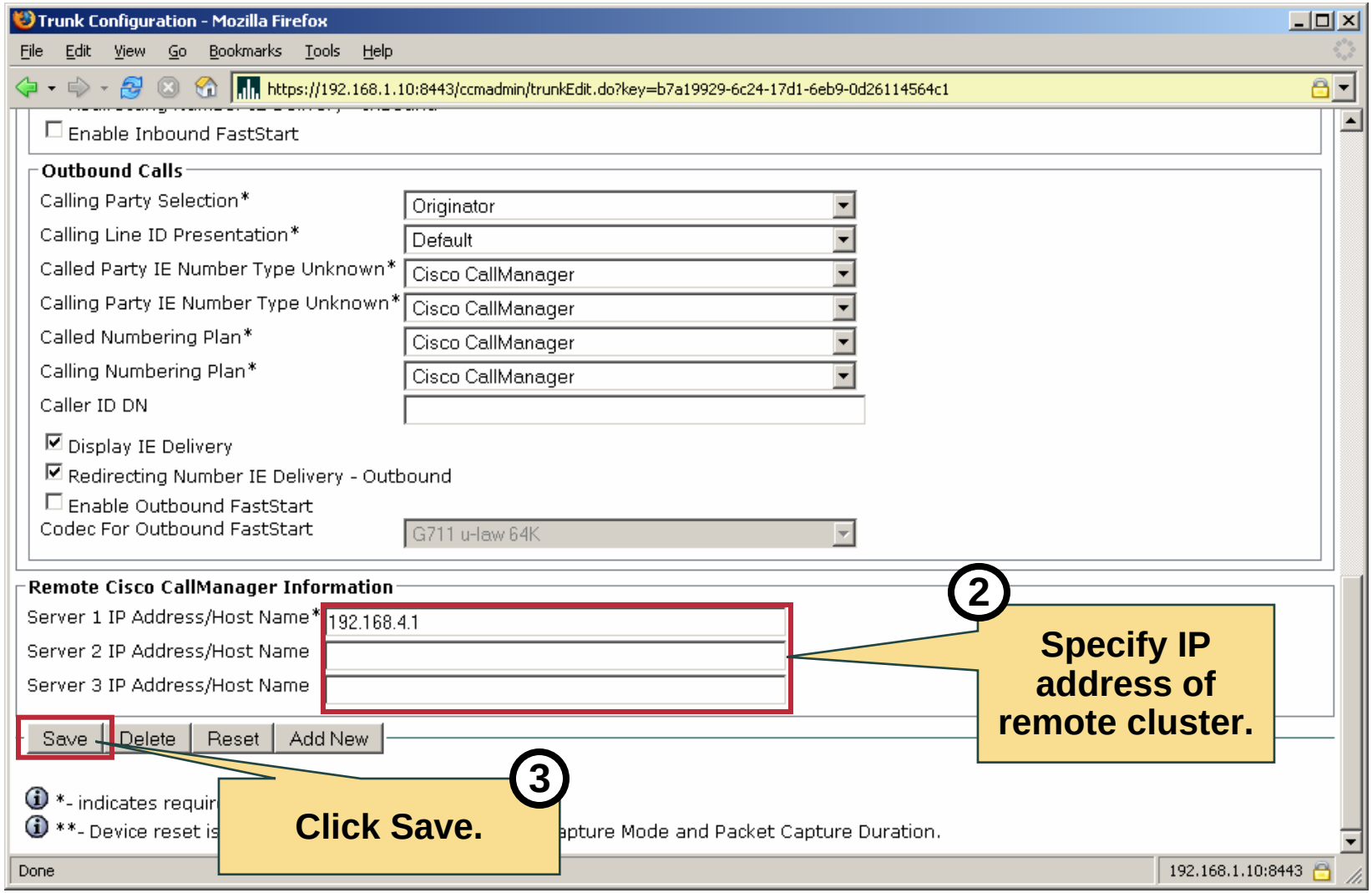
Configuring Non-Gatekeeper Controlled Intercluster Trunks: Configuring Options



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Configuring Gatekeeper Controlled Intercluster Trunks

Gatekeeper Controlled Intercluster trunk configuration steps:

1. **Add Gatekeeper.**
2. **Add Gatekeeper Controlled ICT.**
3. **Specify trunk parameters.**

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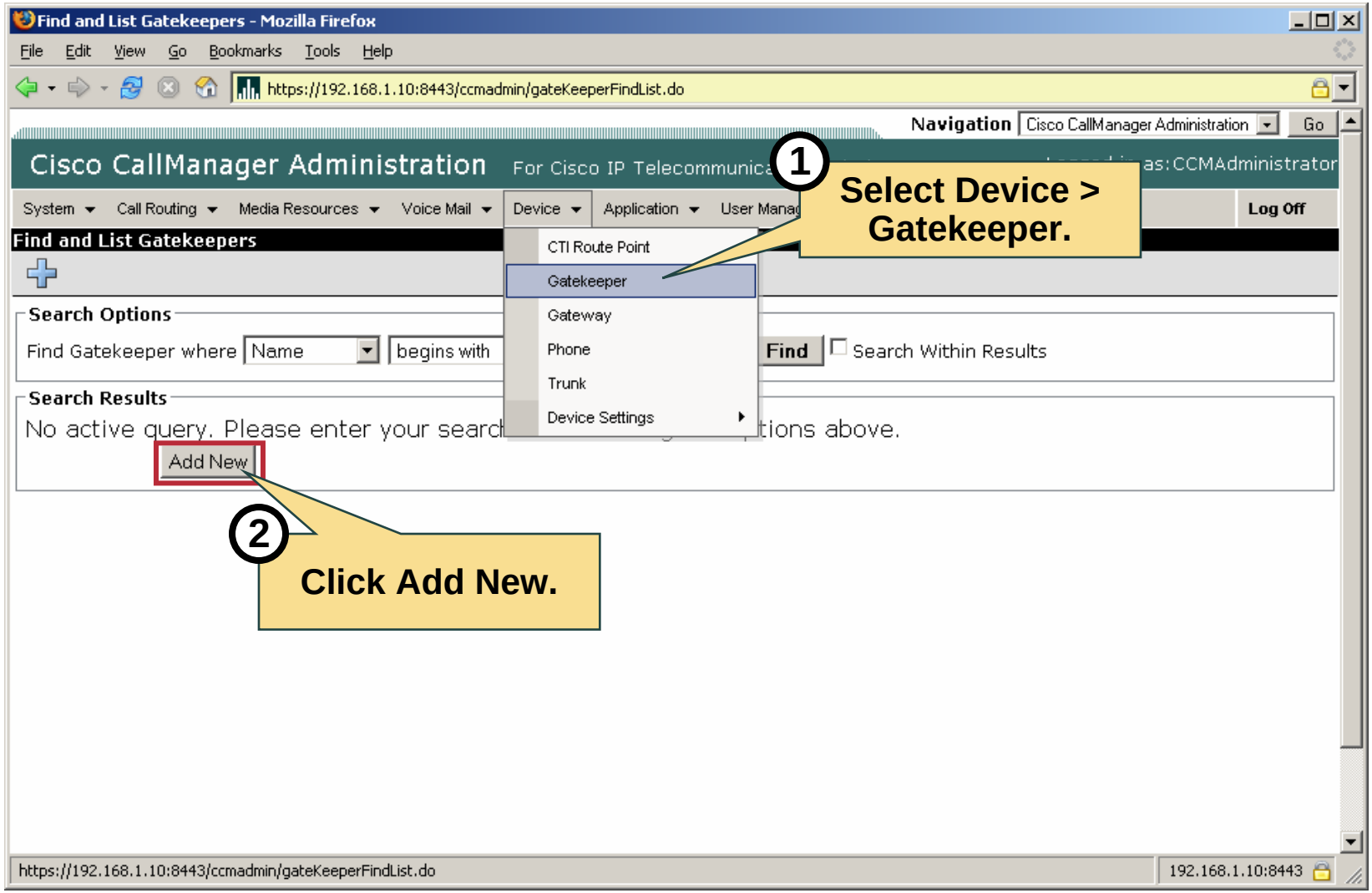
Configuring Gatekeeper Controlled Intercluster Trunks: Add Gatekeeper



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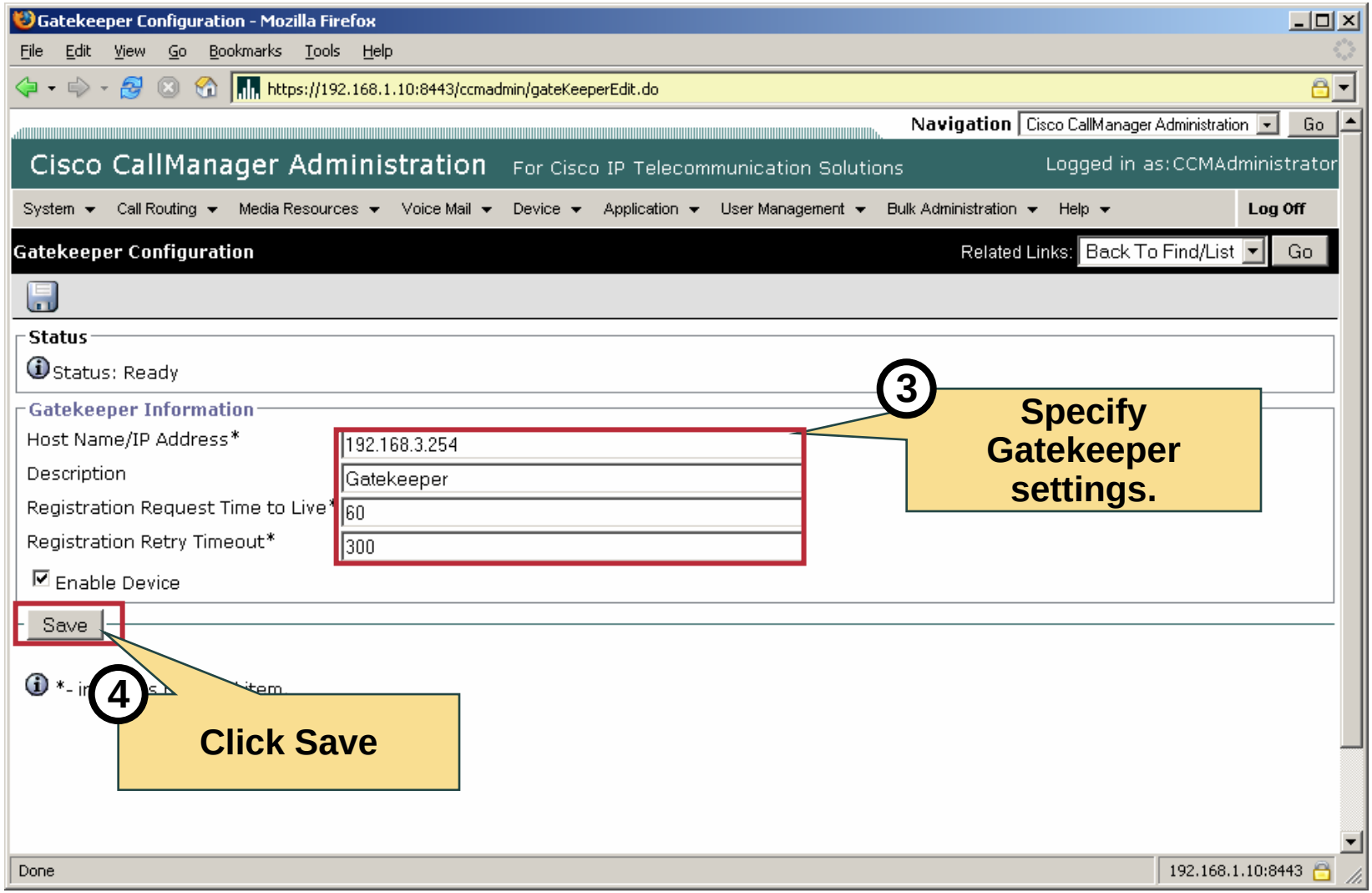
Configuring Gatekeeper Controlled Intercluster Trunks: Add Gatekeeper



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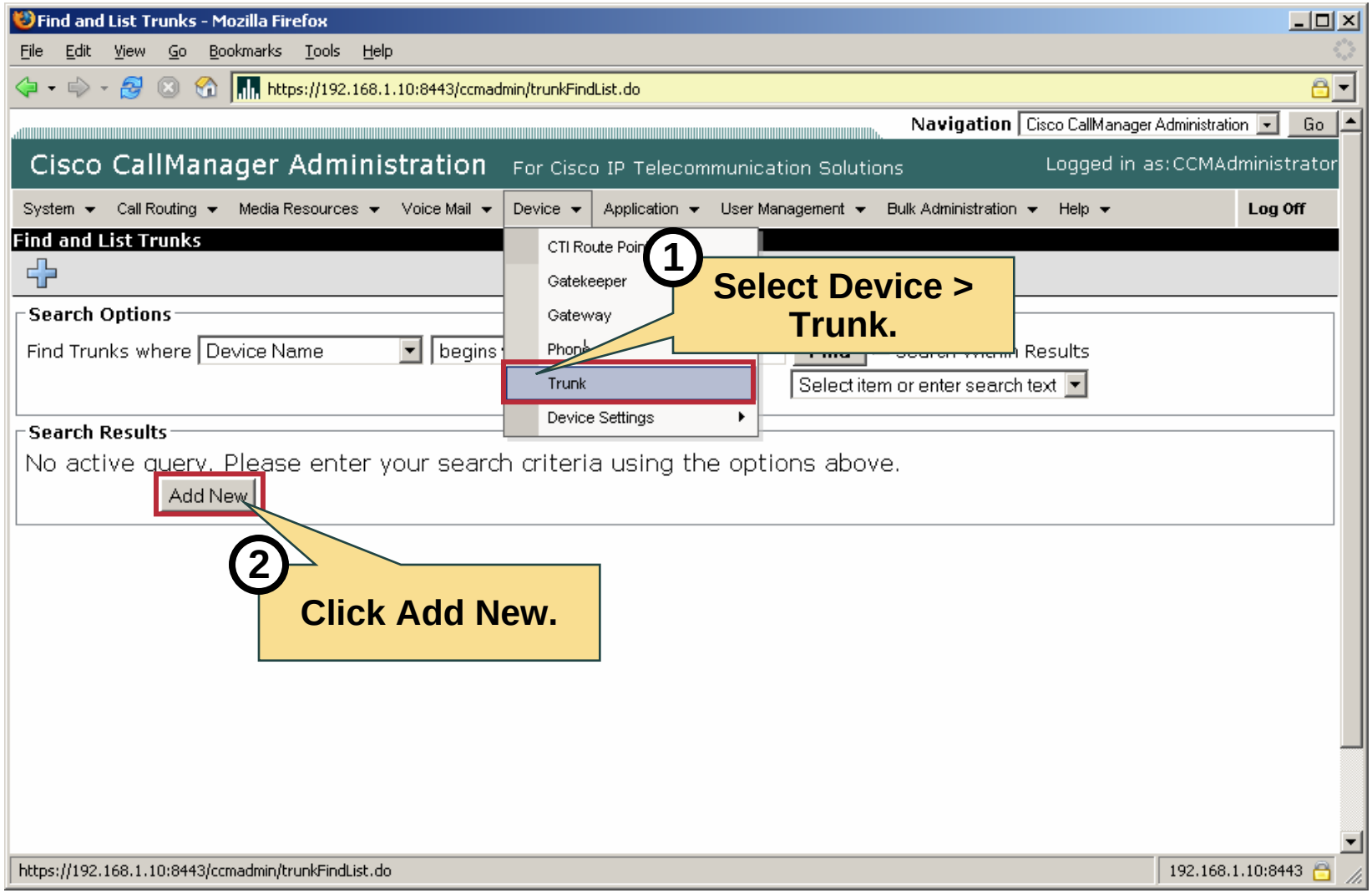
Configuring Gatekeeper Controlled Intercluster Trunks: Add Trunk



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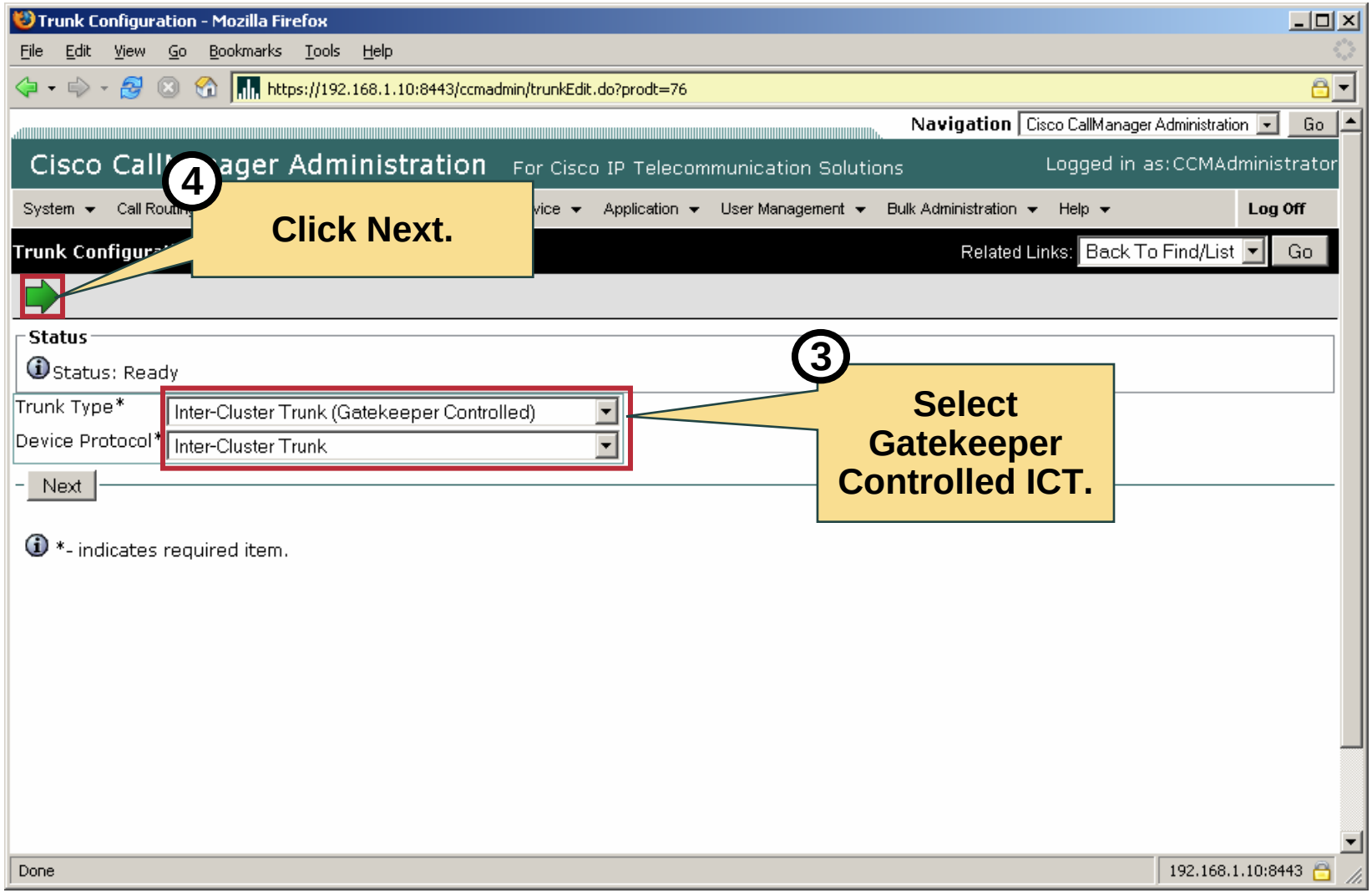
Configuring Gatekeeper Controlled Intercluster Trunks: Add Trunk



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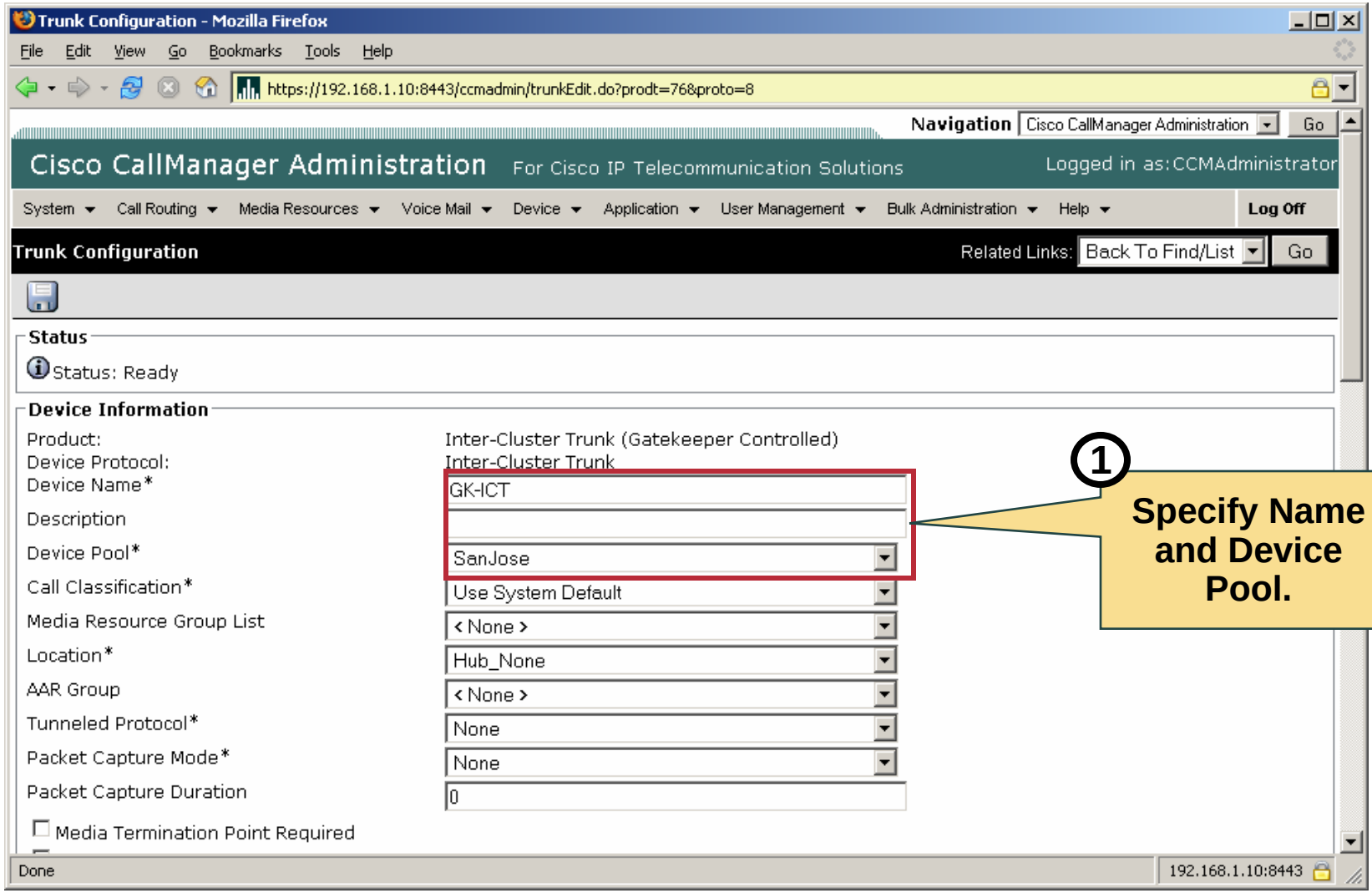
Configuring Gatekeeper Controlled Intercluster Trunks: Configure Trunk



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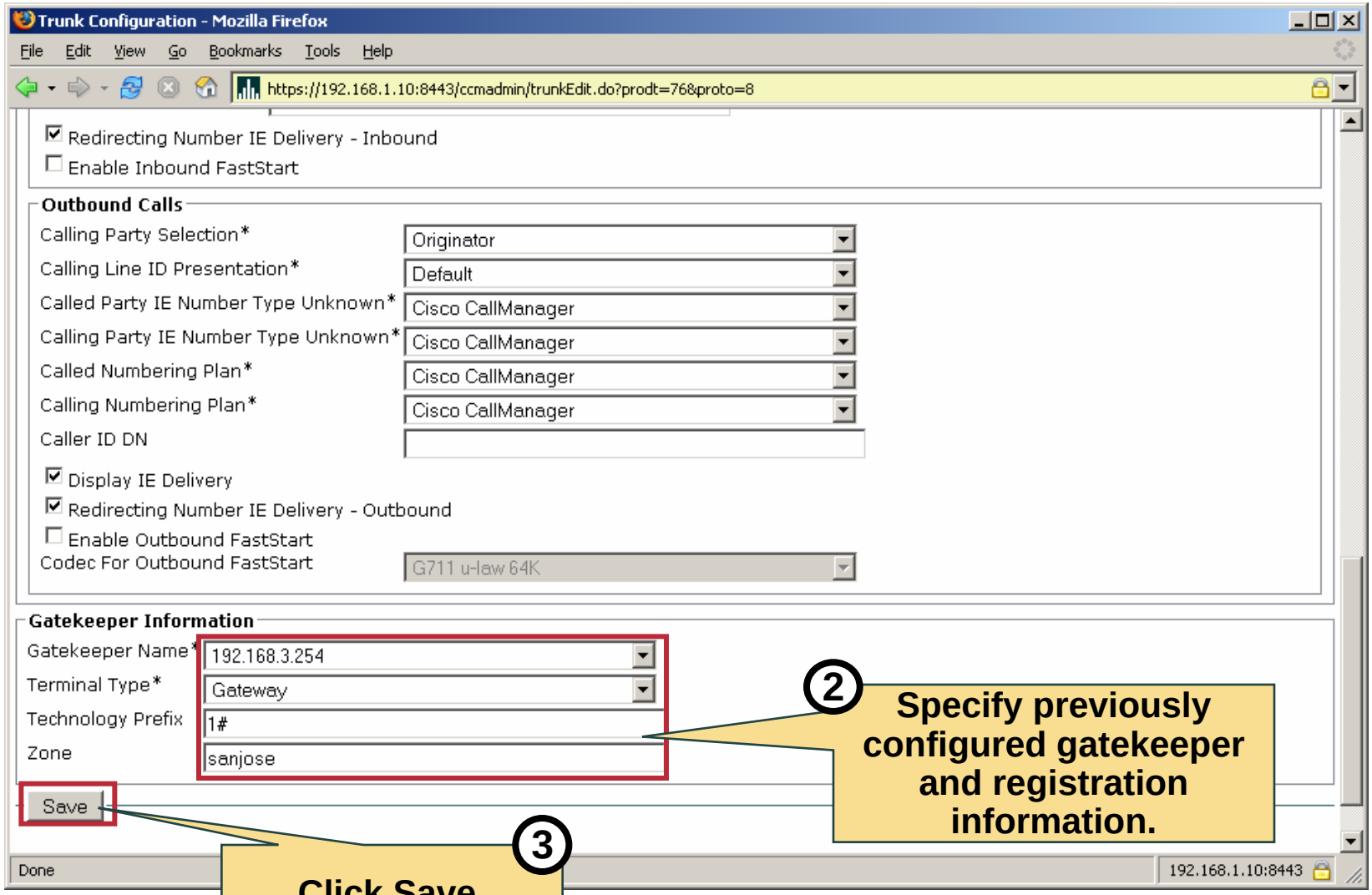
Configuring Gatekeeper Controlled Intercluster Trunks



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Configuring QSIG for Trunks

Any Inter-Cluster Trunk can be configured for QSIG:

- **Done via trunk configuration settings.**
- **Can change existing trunks to QSIG.**

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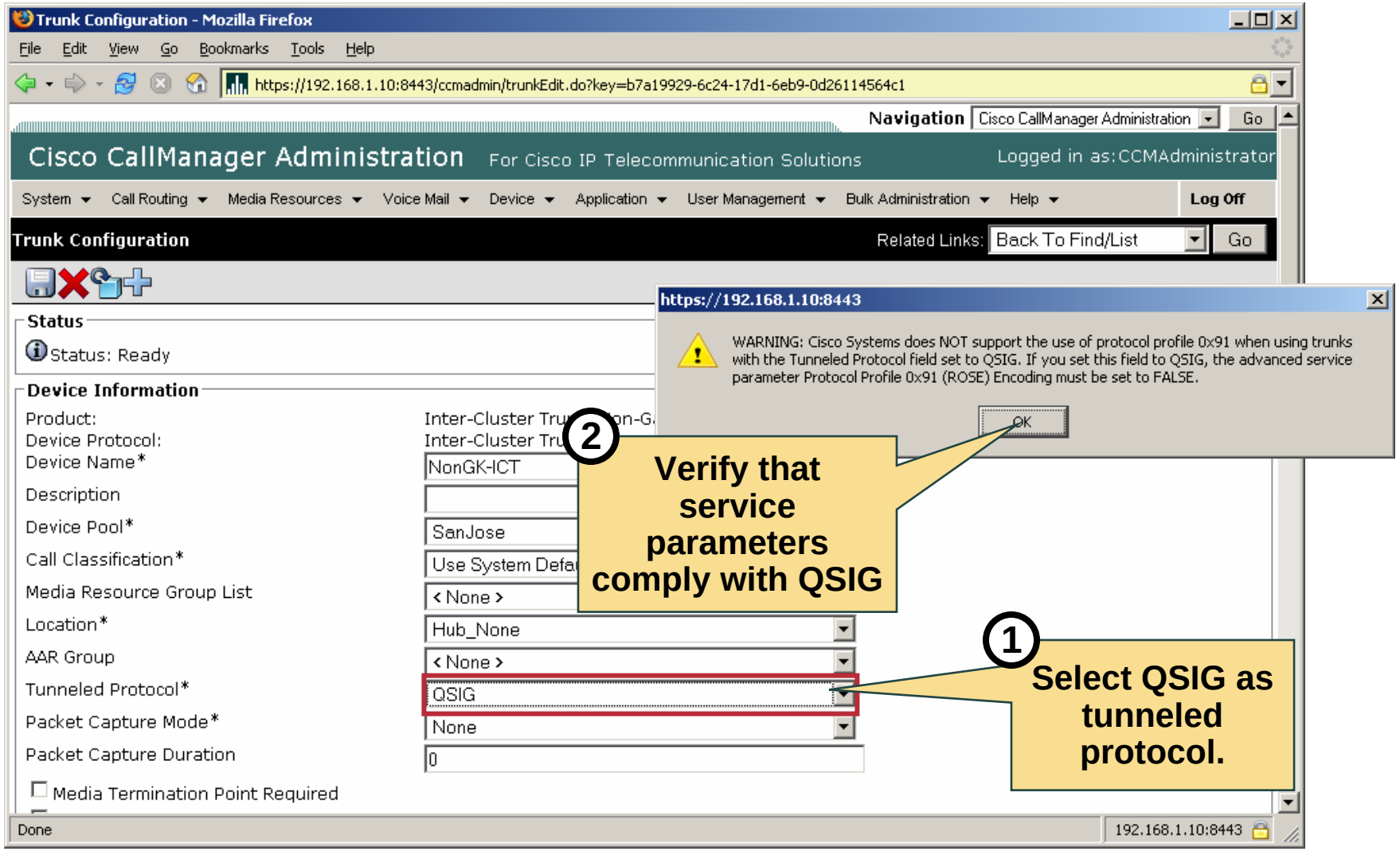
Configuring QSIG for Trunks



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Configuring SIP Trunks

SIP trunk configuration steps:

1. **Add SIP trunk.**
2. **Specify trunk parameters.**

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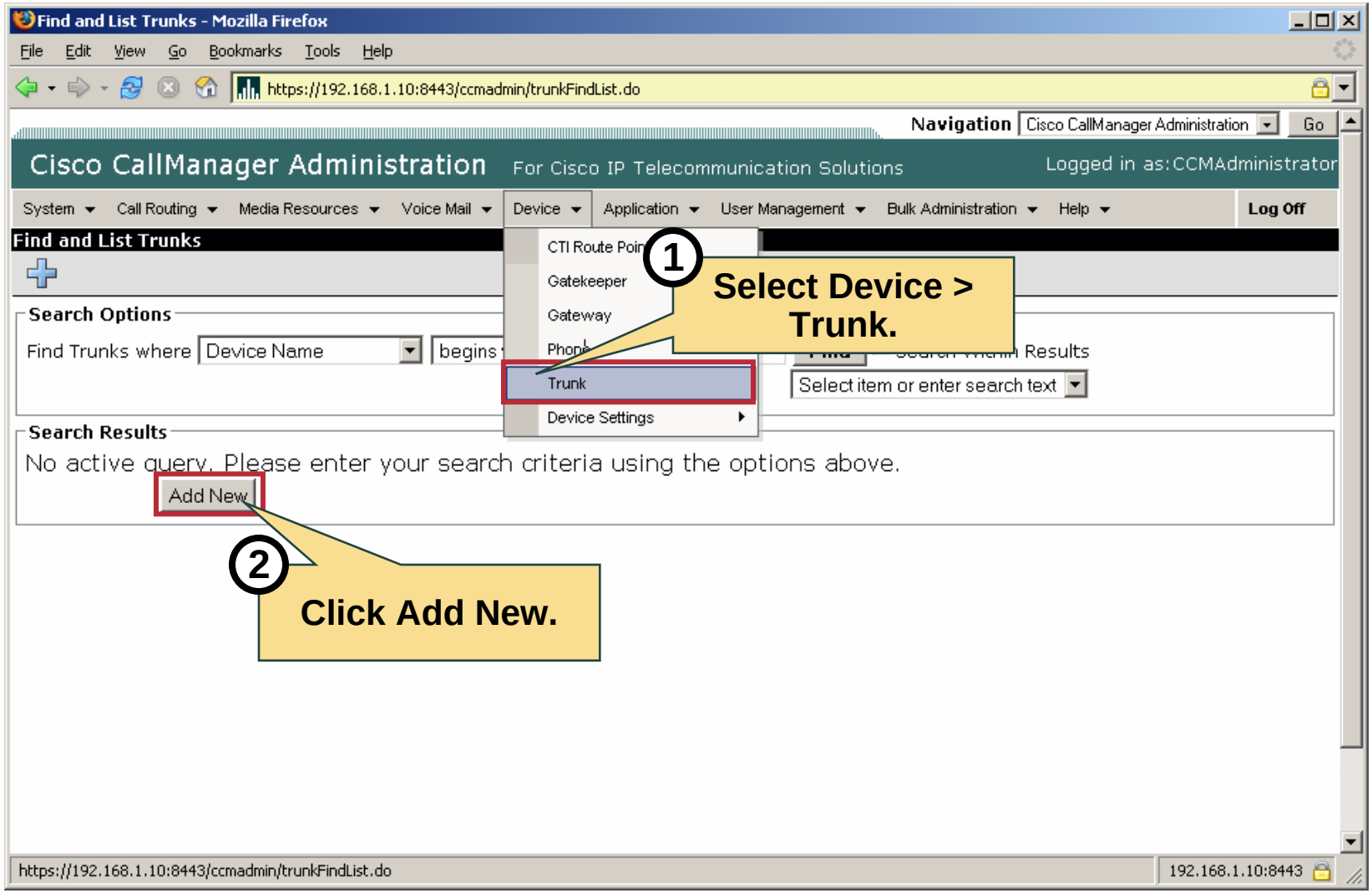
Configuring SIP Trunks: Add Trunk



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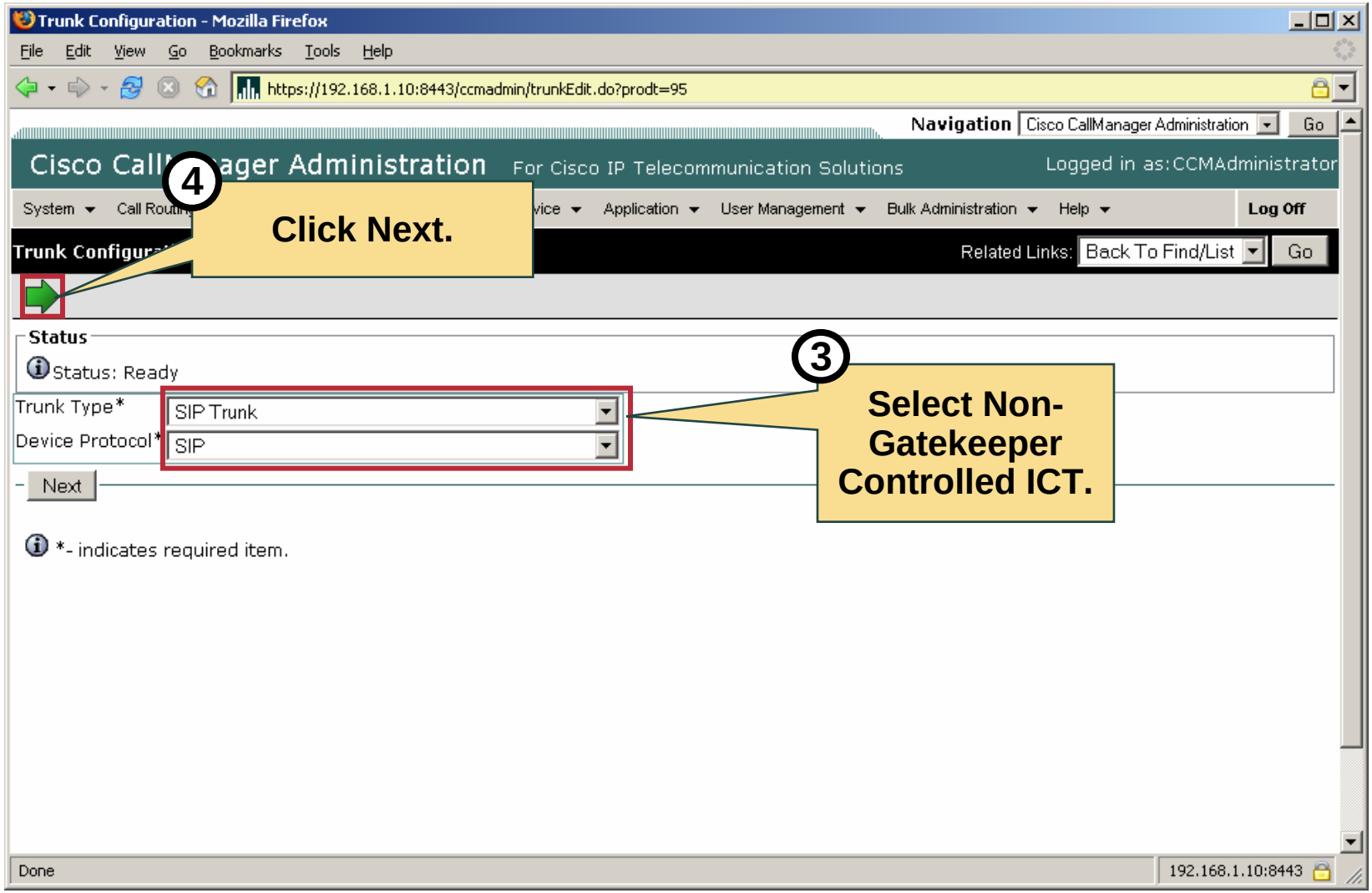
Configuring SIP Trunks: Add Trunk



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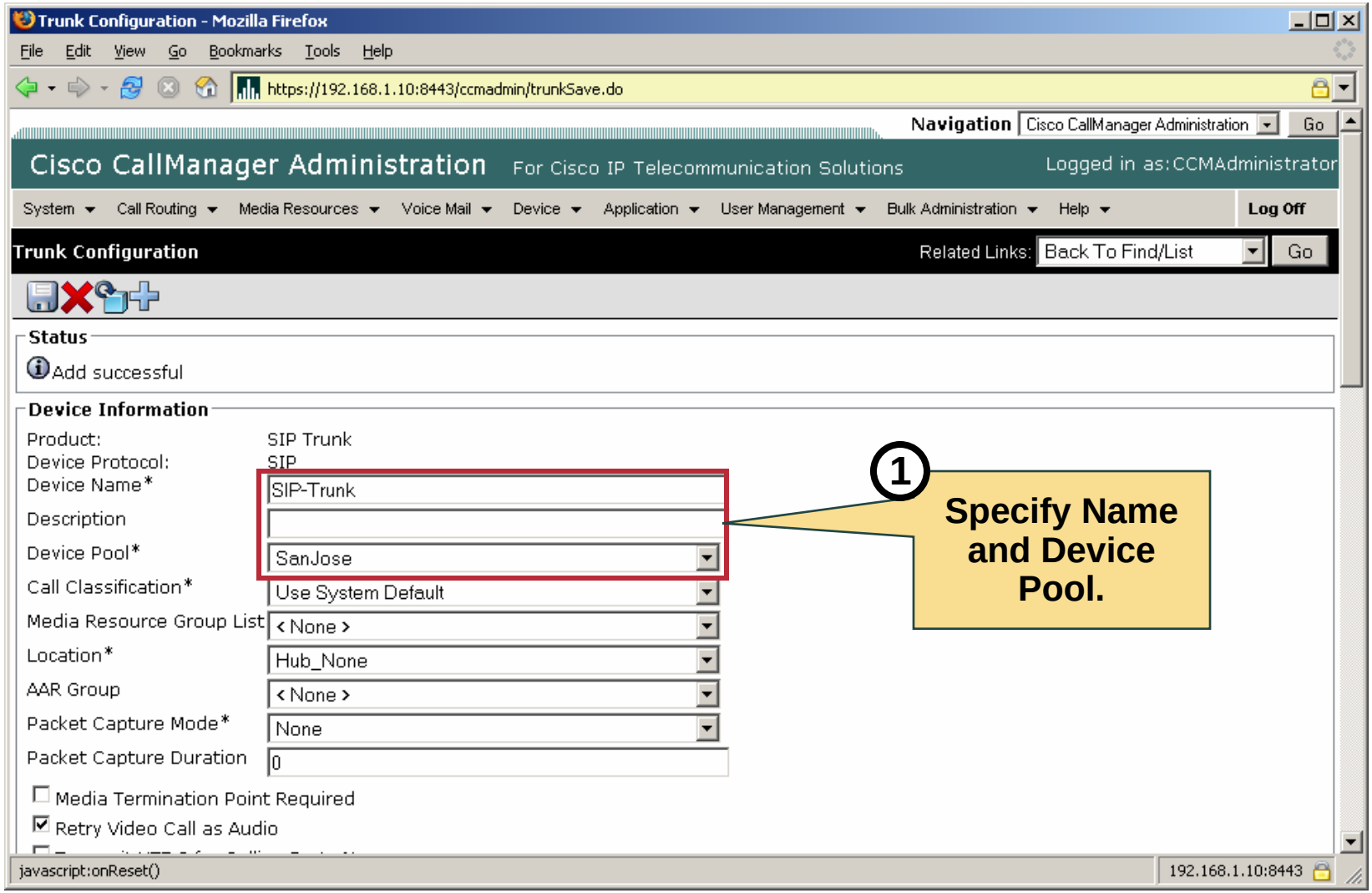
Configuring SIP Trunks: Configure Trunk



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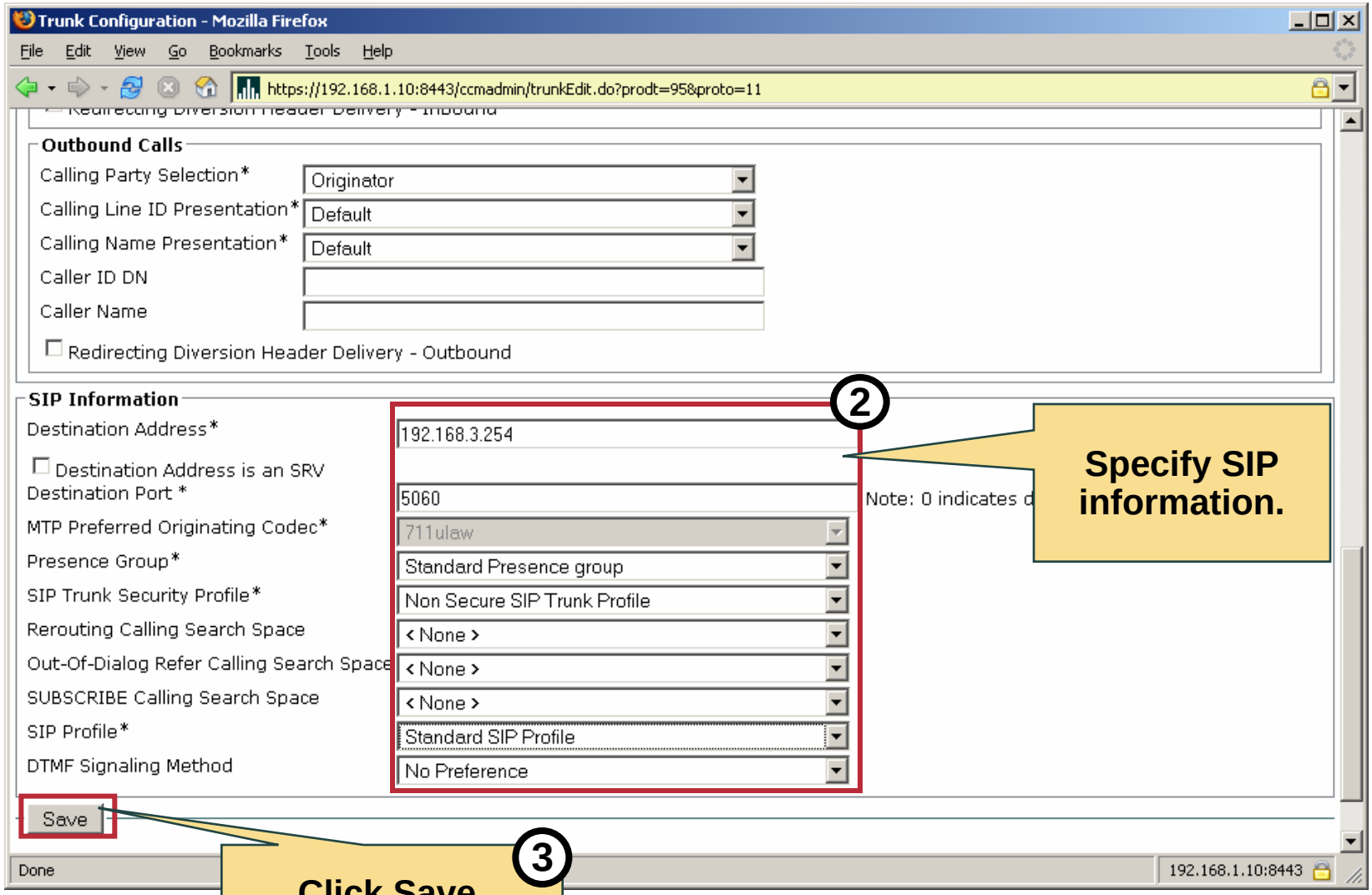
Configuring SIP Trunks: Configure Trunk



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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.

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Deploying a Dial Plan on CallManager 5.0

Configuring a Dial Plan

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Objectives

- **Dial Plan Scenario**
- **Configuring Intrasite Routing**
- **Configuring On-Cluster Intersite Routing**
- **Configuring Off-Cluster Intersite Routing**
- **Configuring PSTN Routing and Calling Privileges**

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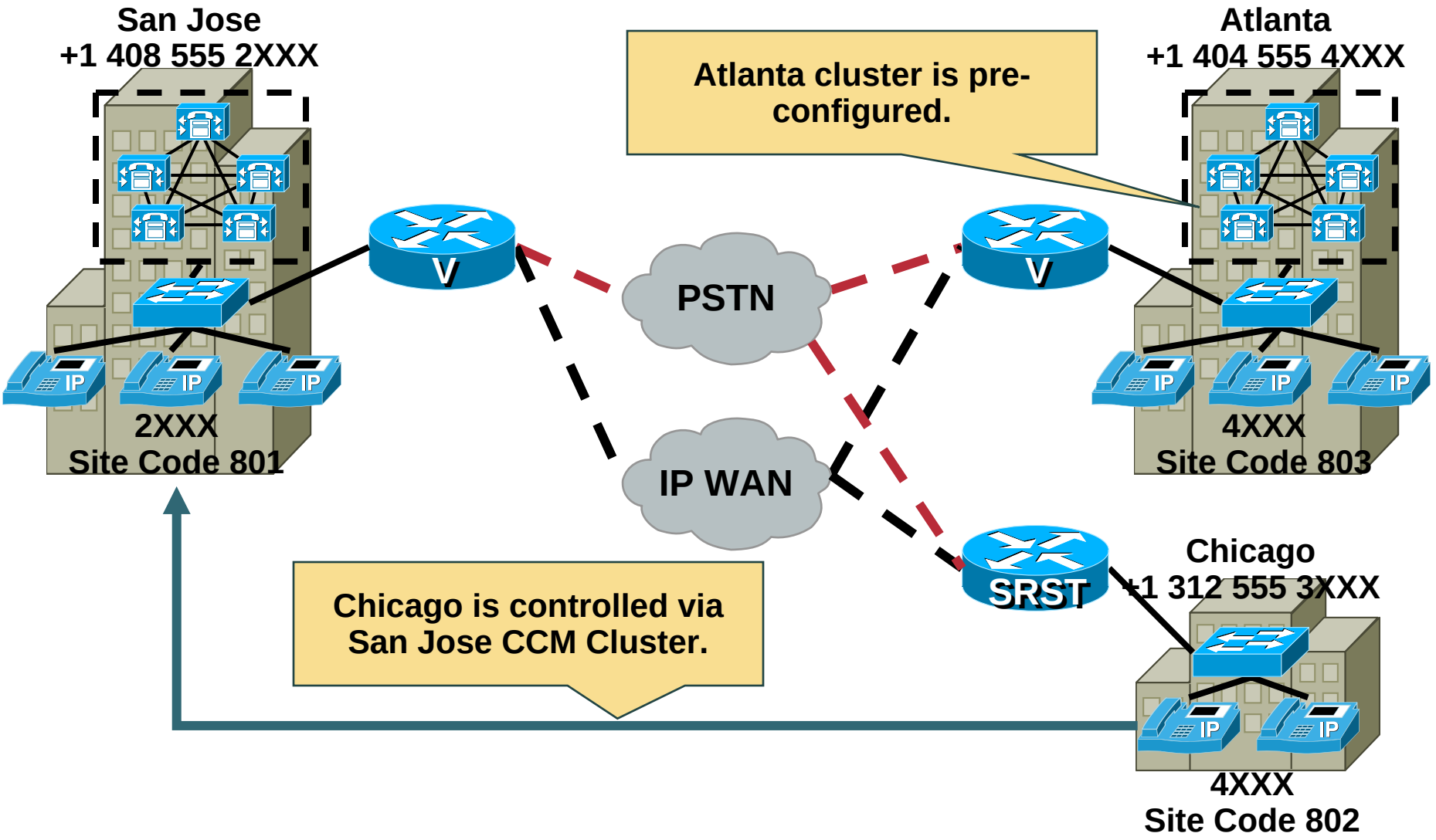
Dial Plan Scenario



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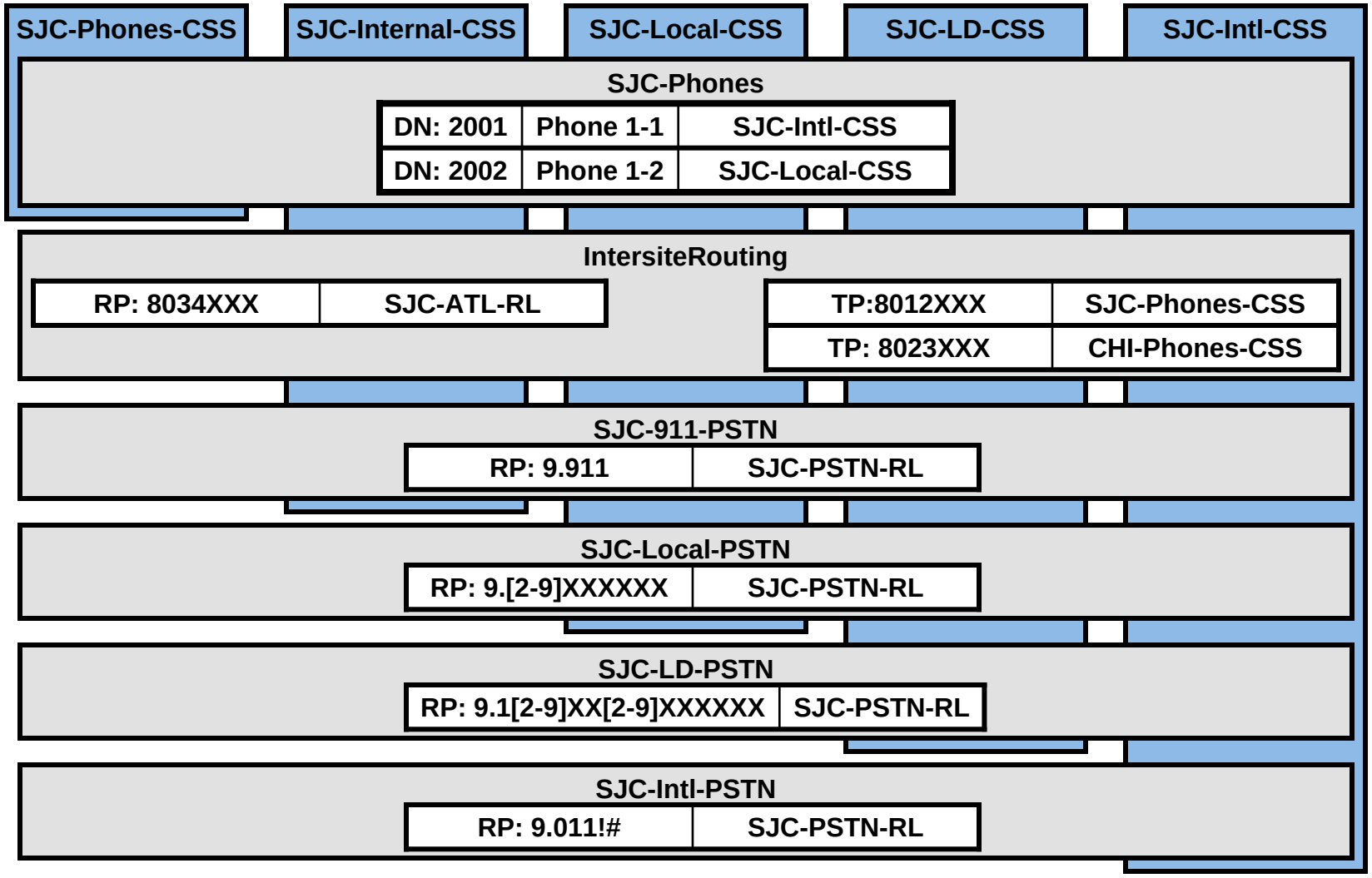
Dial Plan Scenario: San Jose Partitions and CSSs



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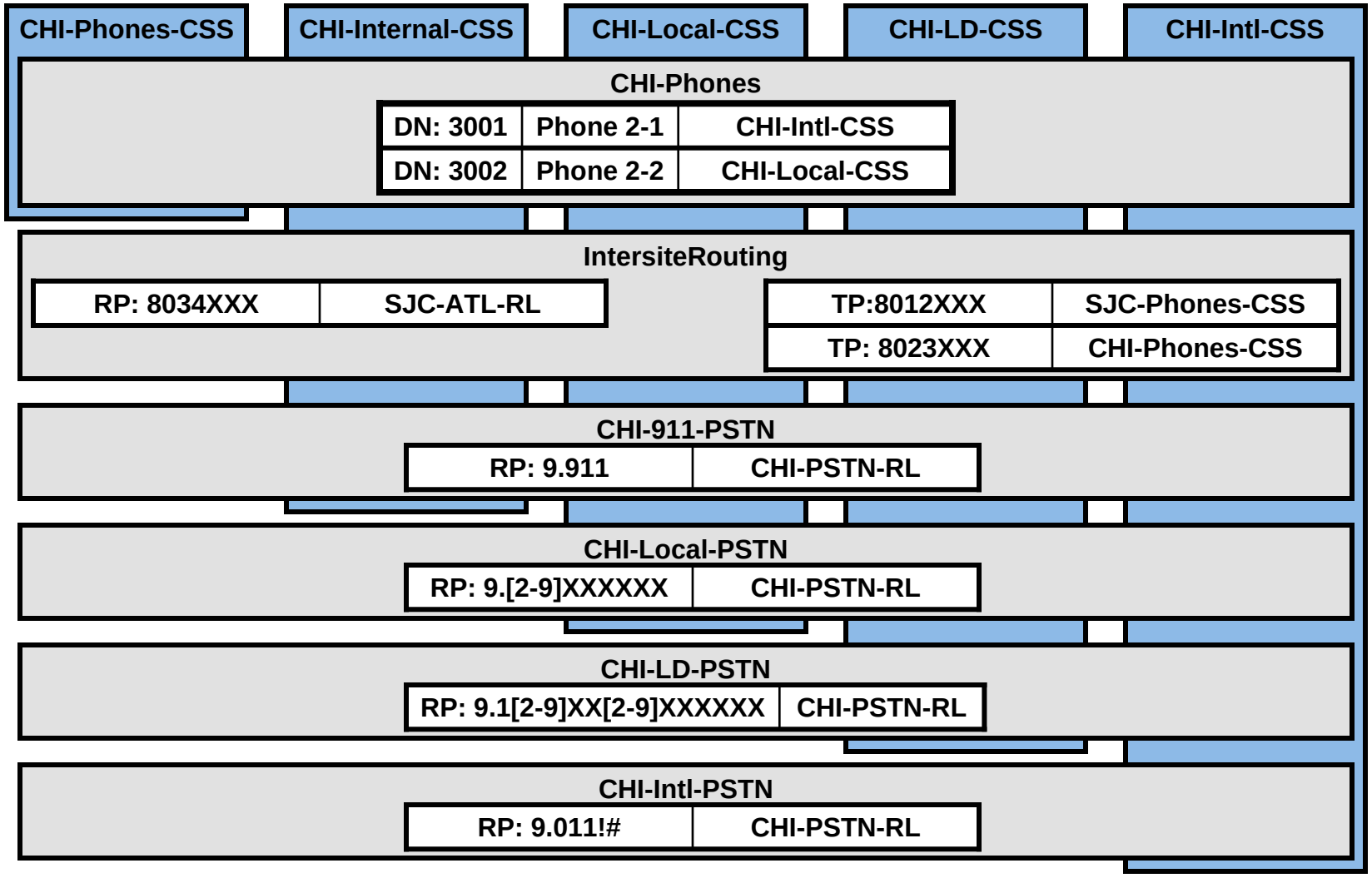
Dial Plan Scenario: Chicago Partitions and CSSs



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Dial Plan Scenario: RPs and TPs

Pattern	Partition	Type	Destination	Digit Manipulation	Description
Intersite Routing					
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
PSTN Routing					
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

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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

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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.**

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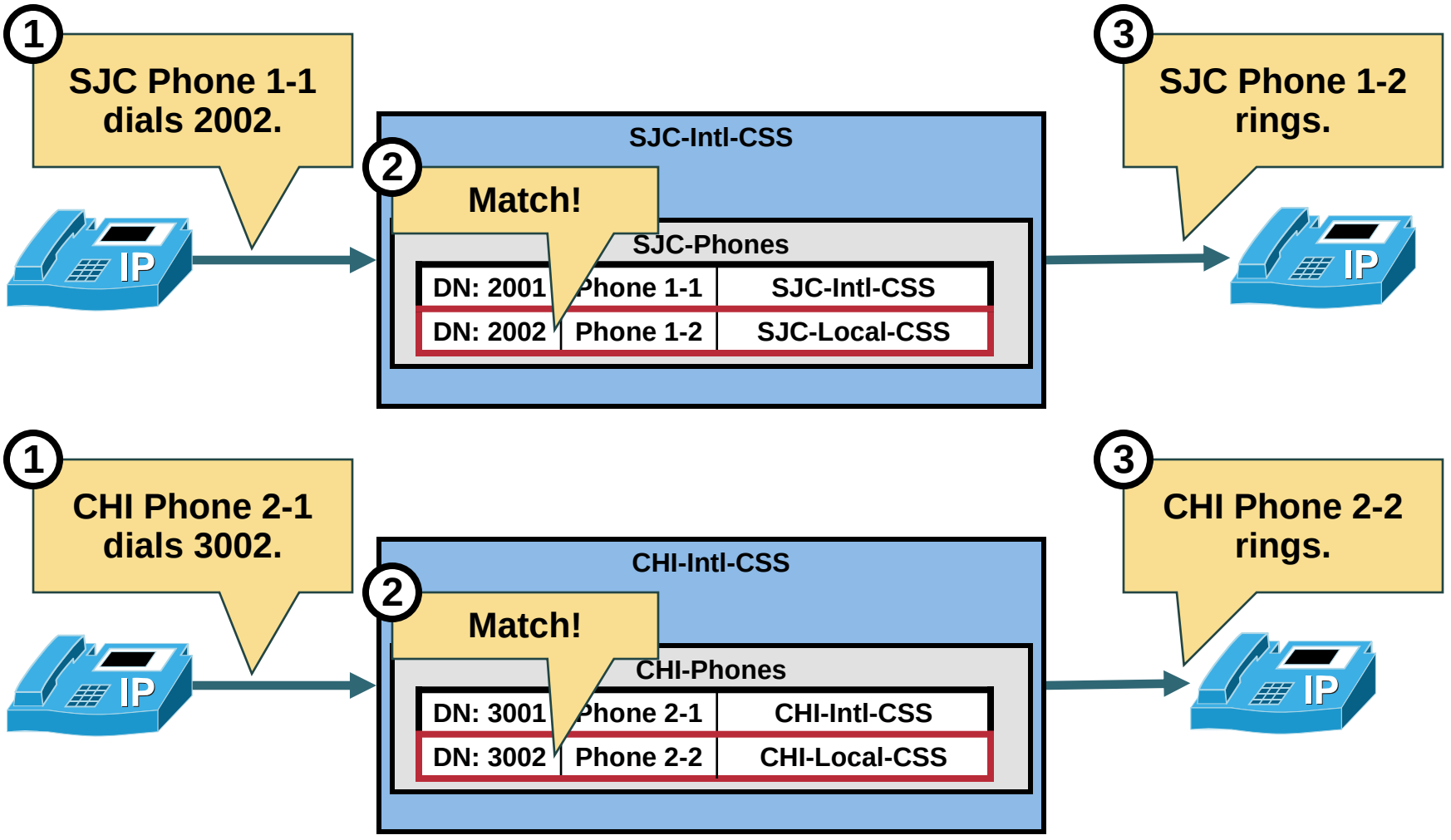
Configuring Intrasite Routing



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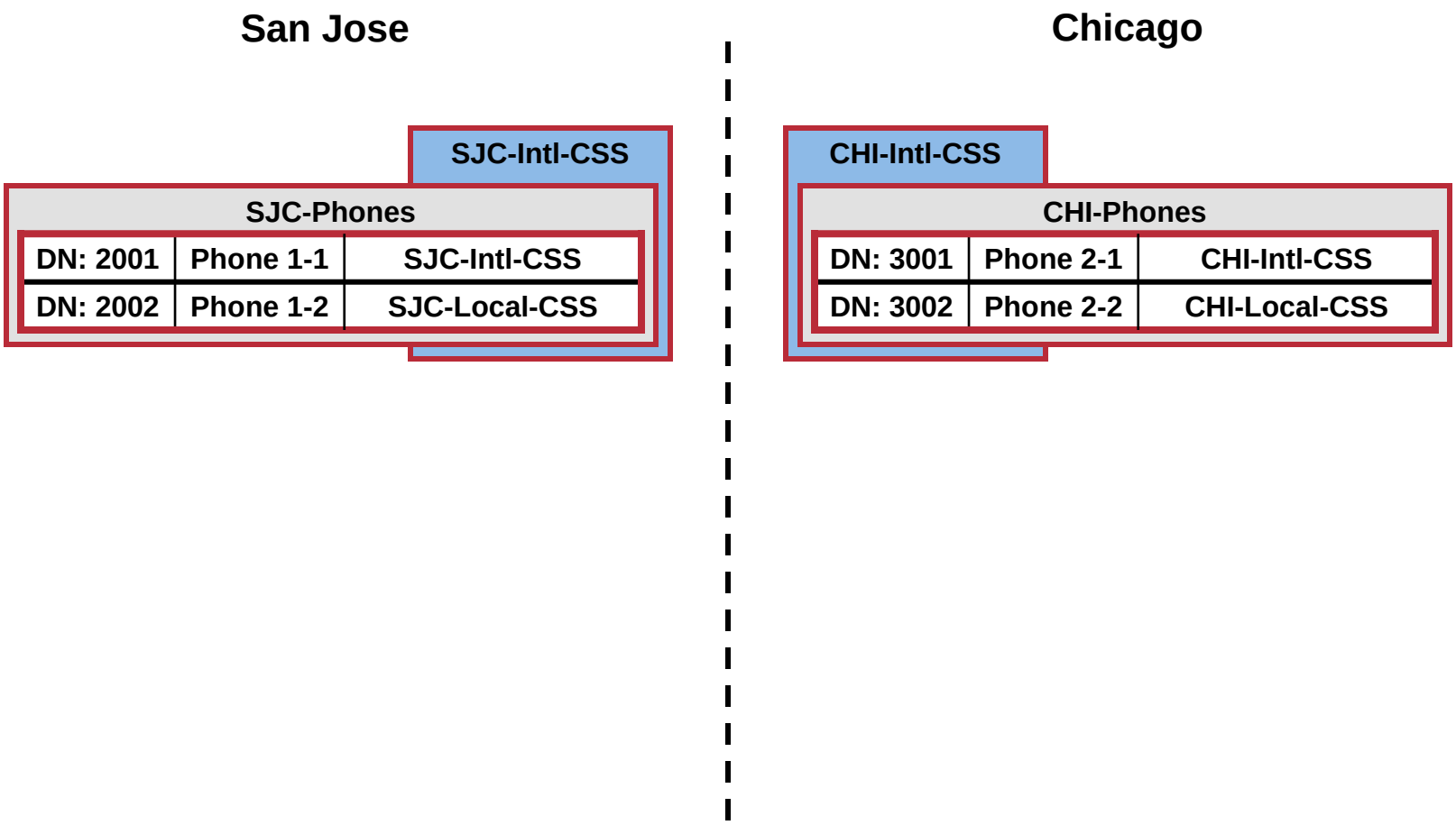
Configuring Intrasite Routing



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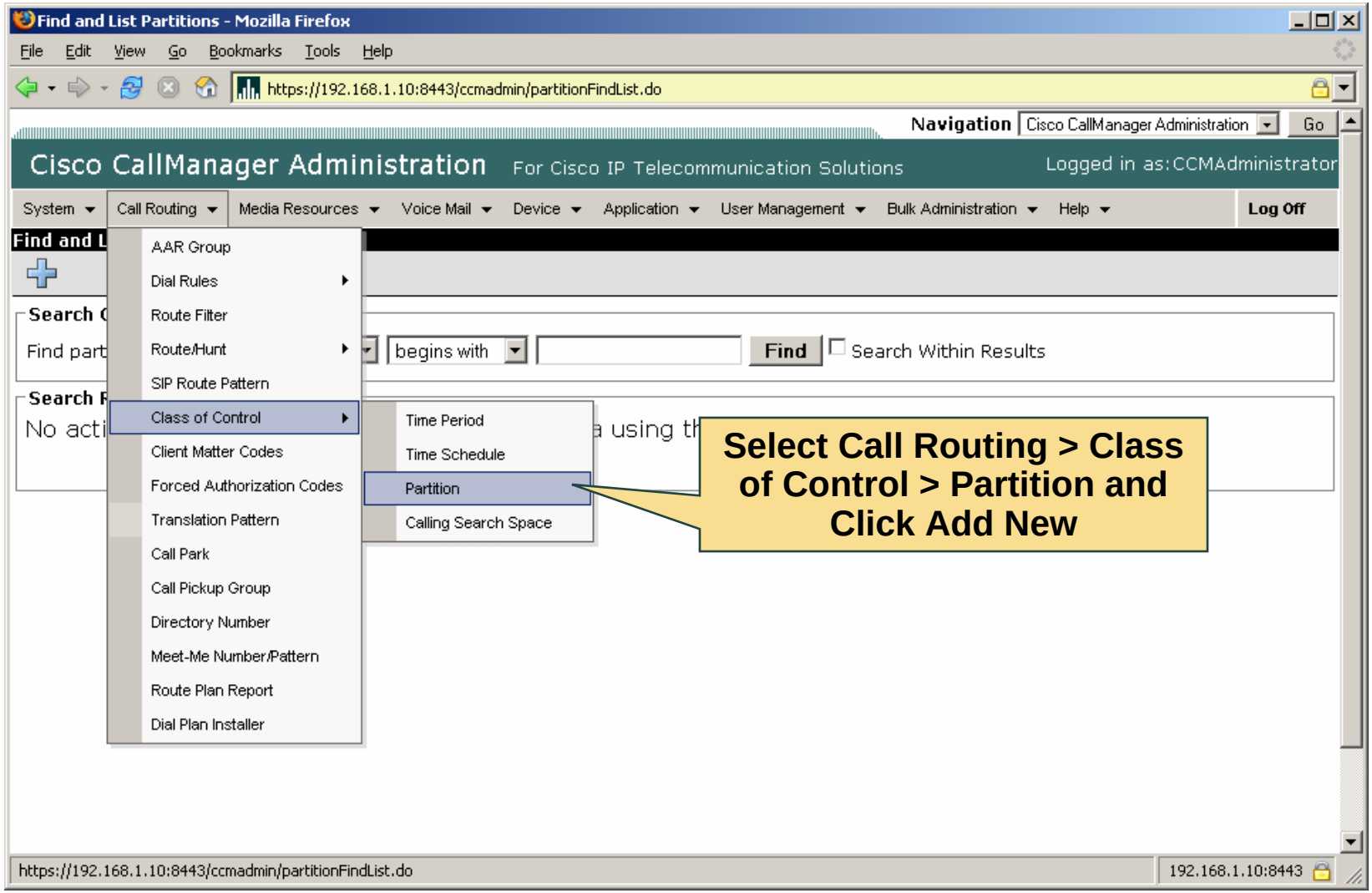
Configuring Intrasite Routing



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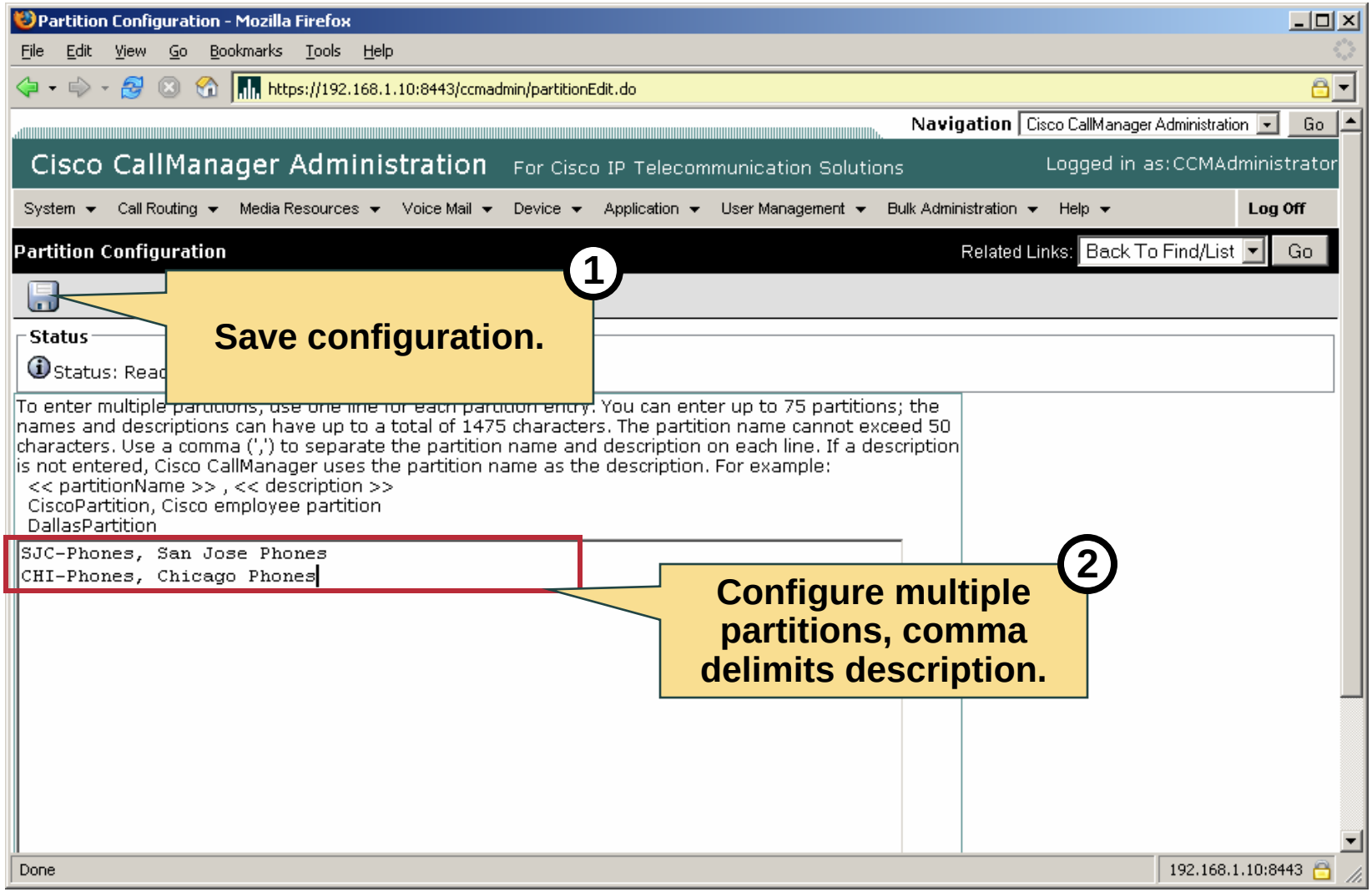
Configuring Intrasite Routing



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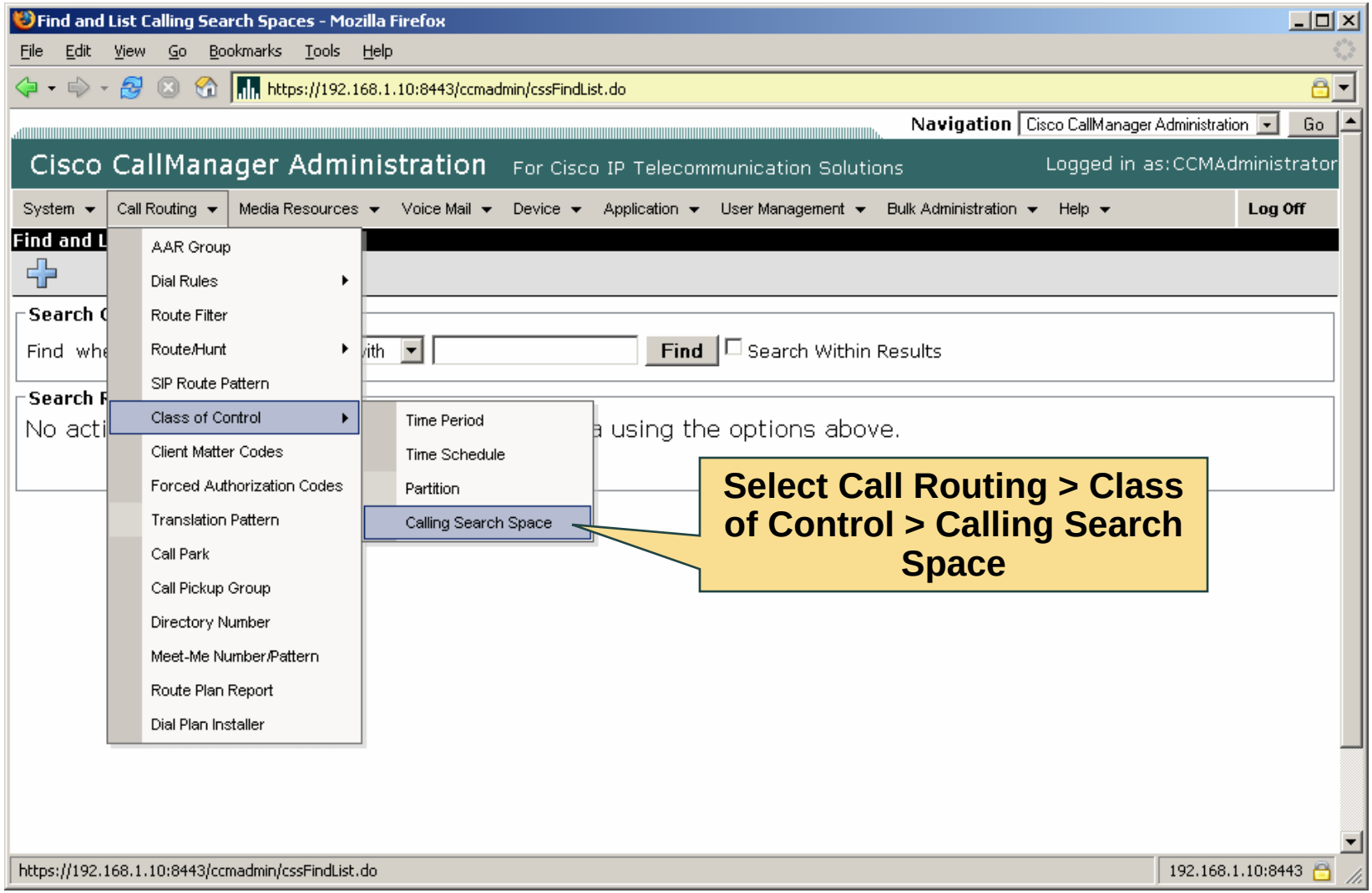
Configuring Intrasite Routing



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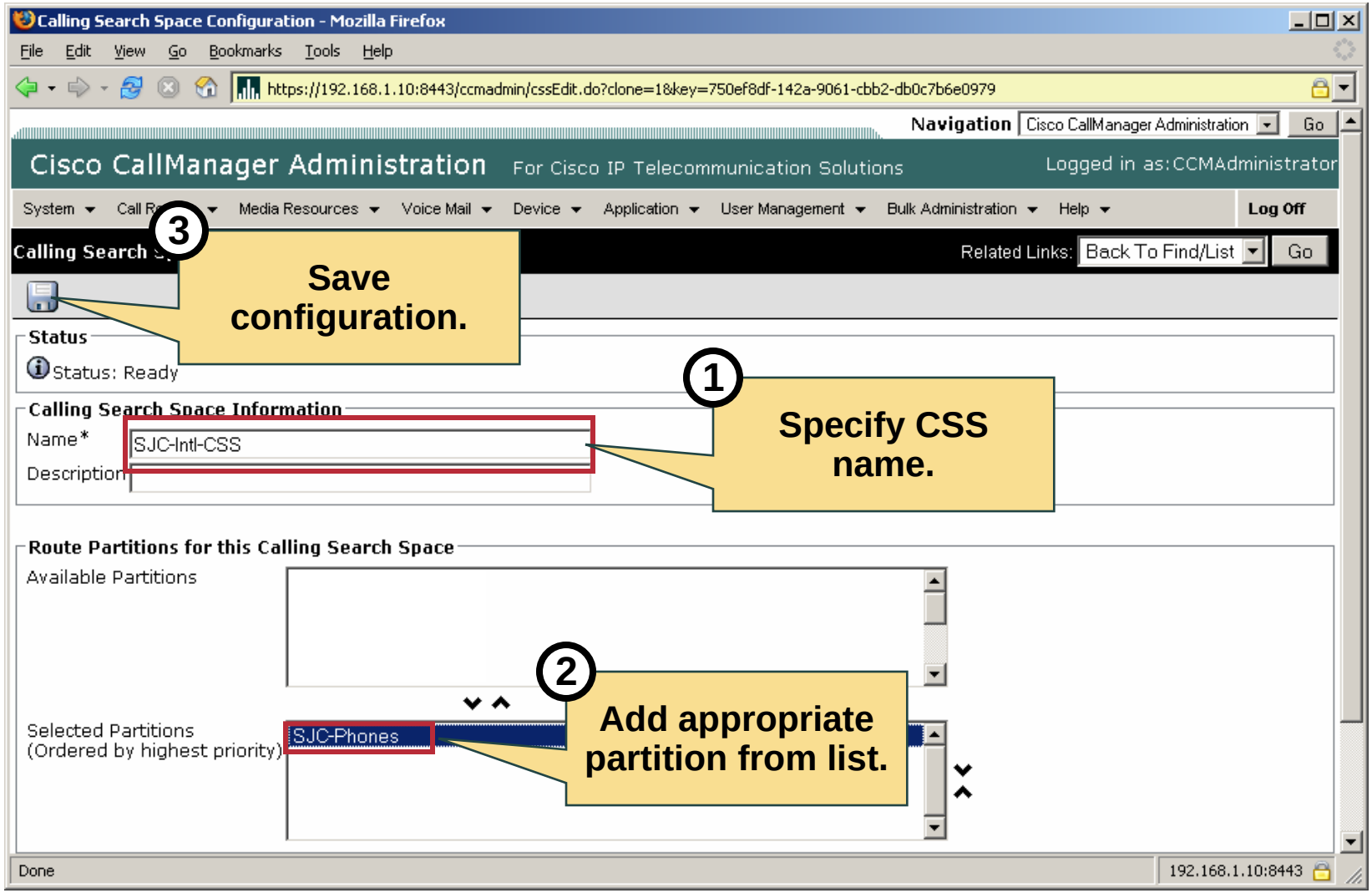
Configuring Intrasite Routing



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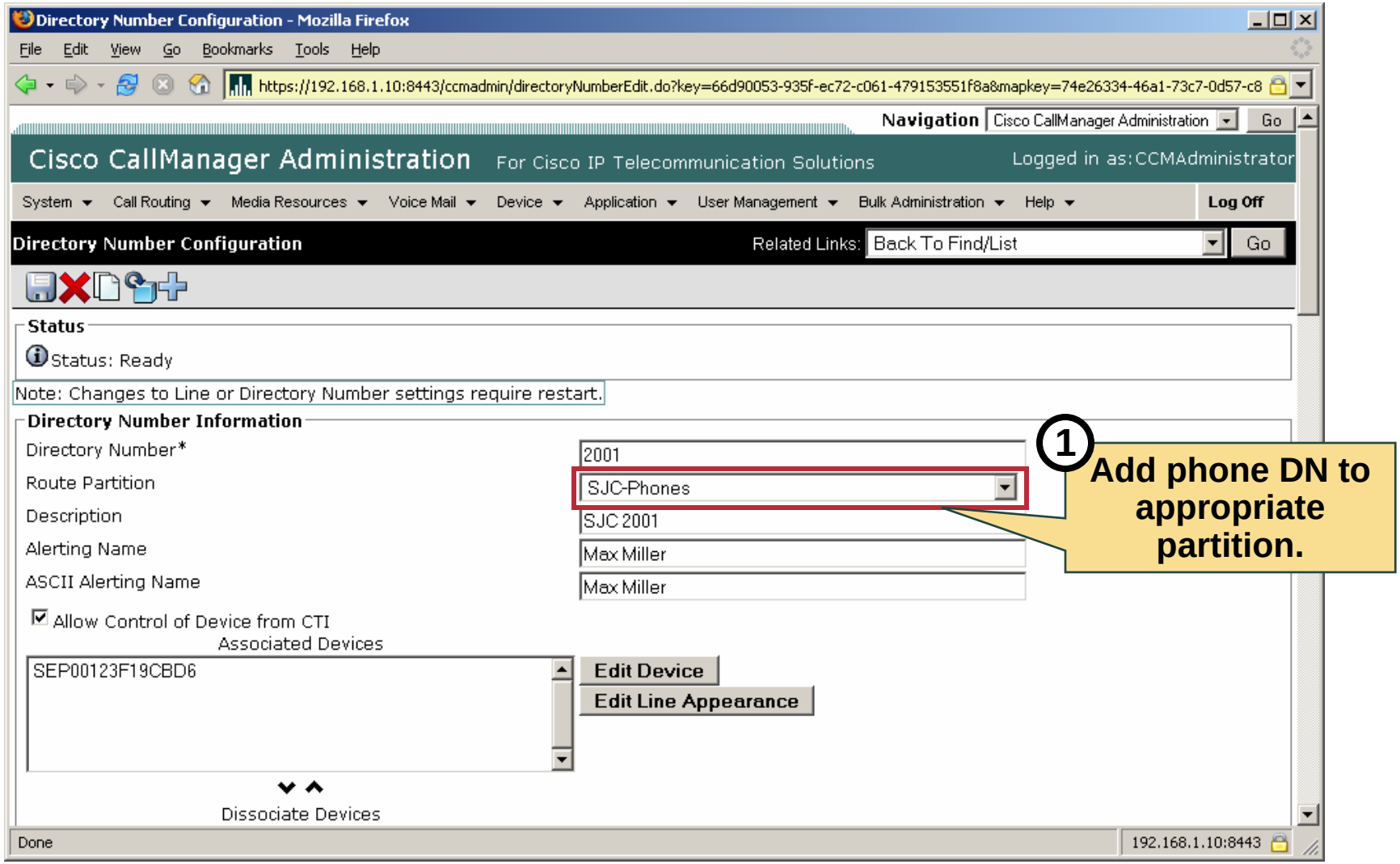
Configuring Intrasite Routing



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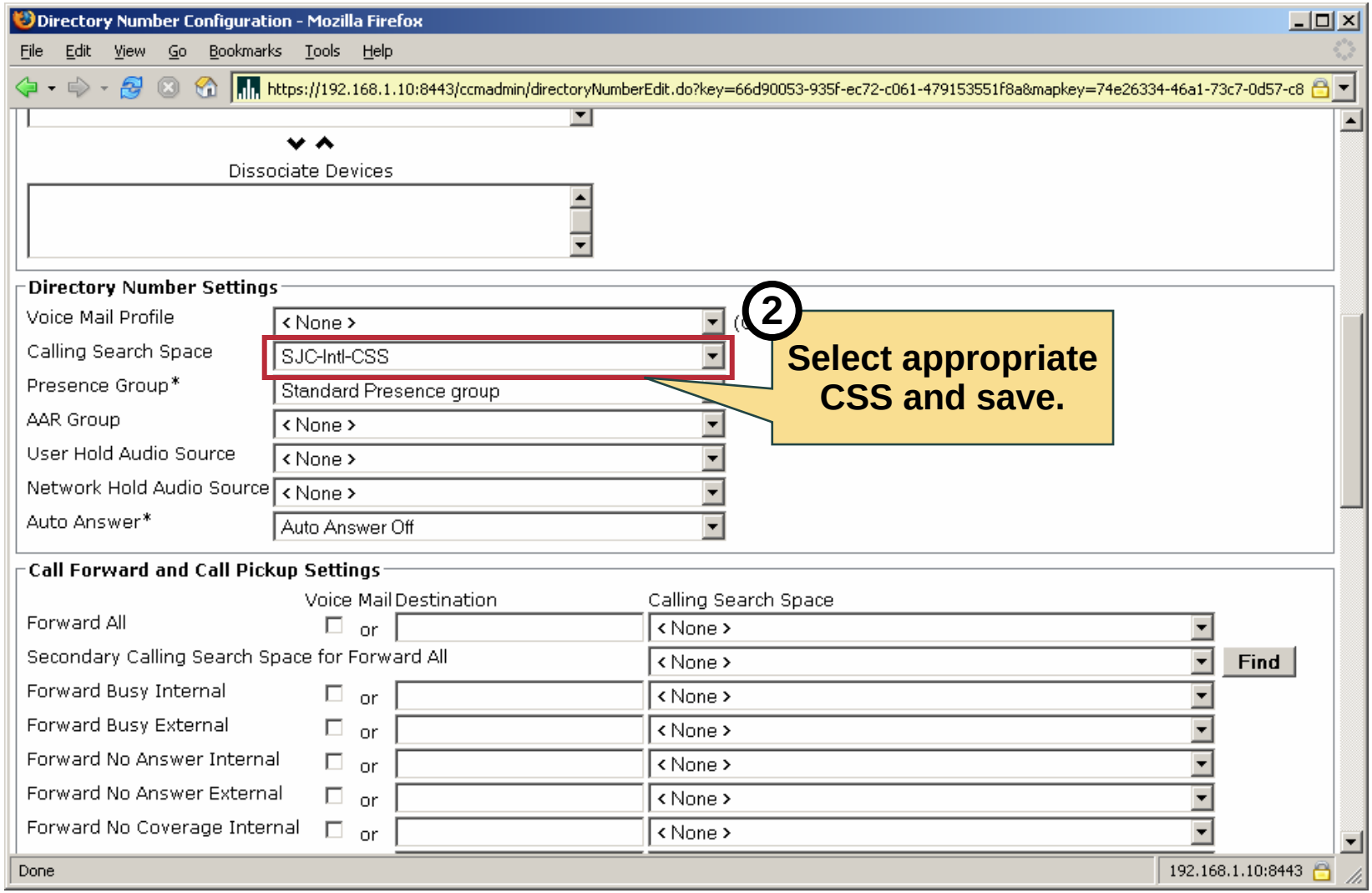
Configuring Intrasite Routing



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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.

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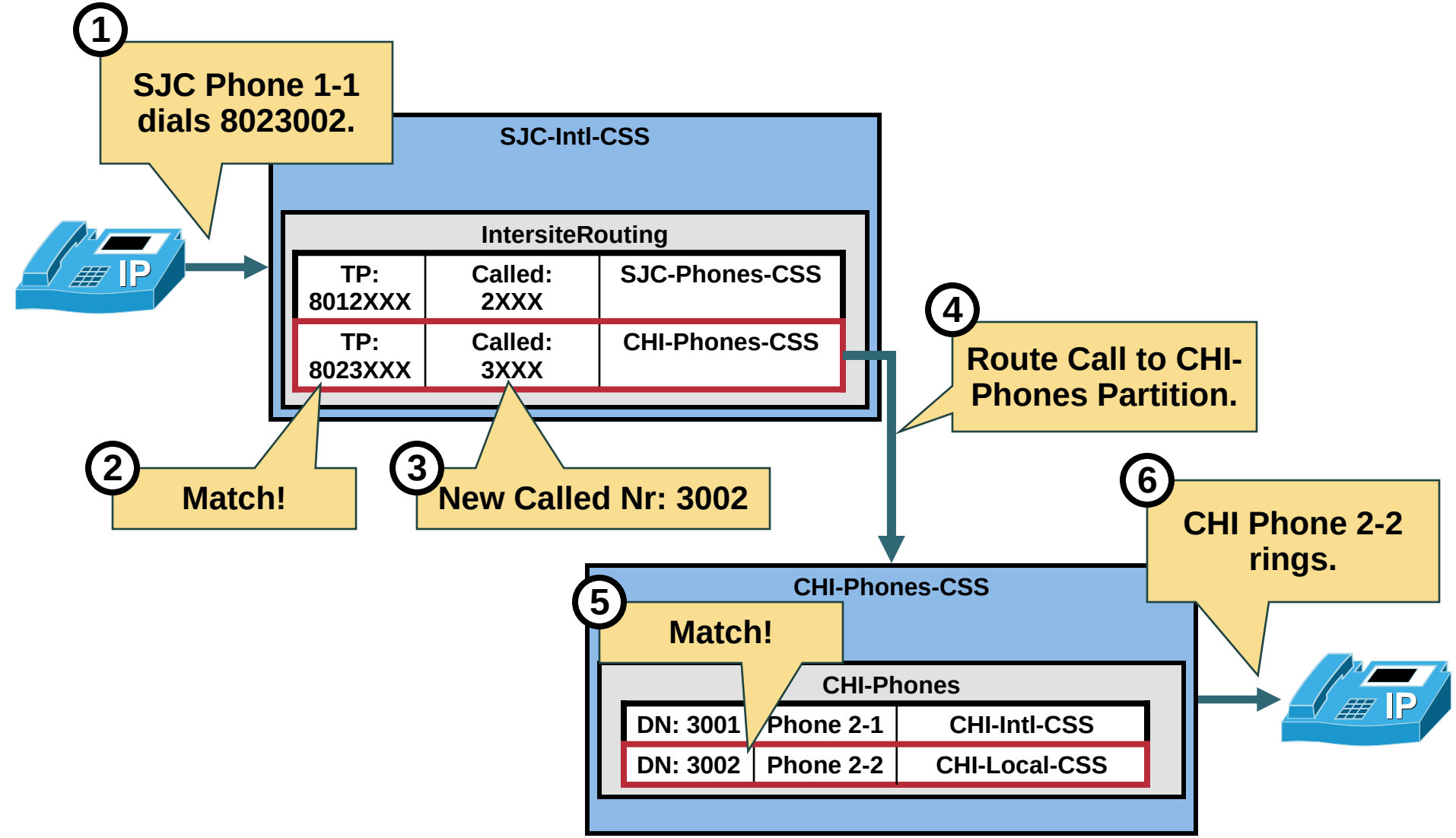
Configuring On-Cluster Intersite Routing



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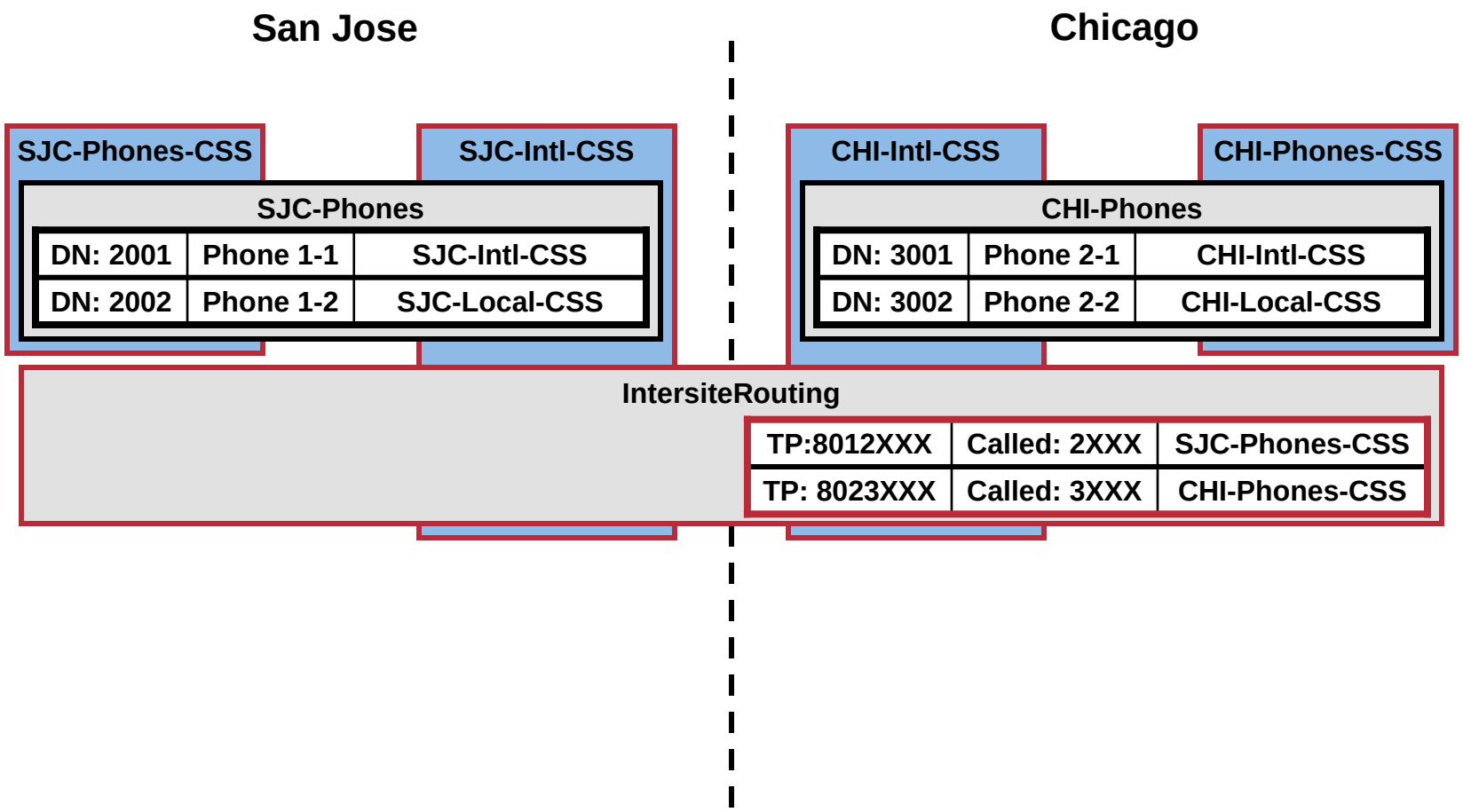
Configuring On-Cluster Intersite Routing



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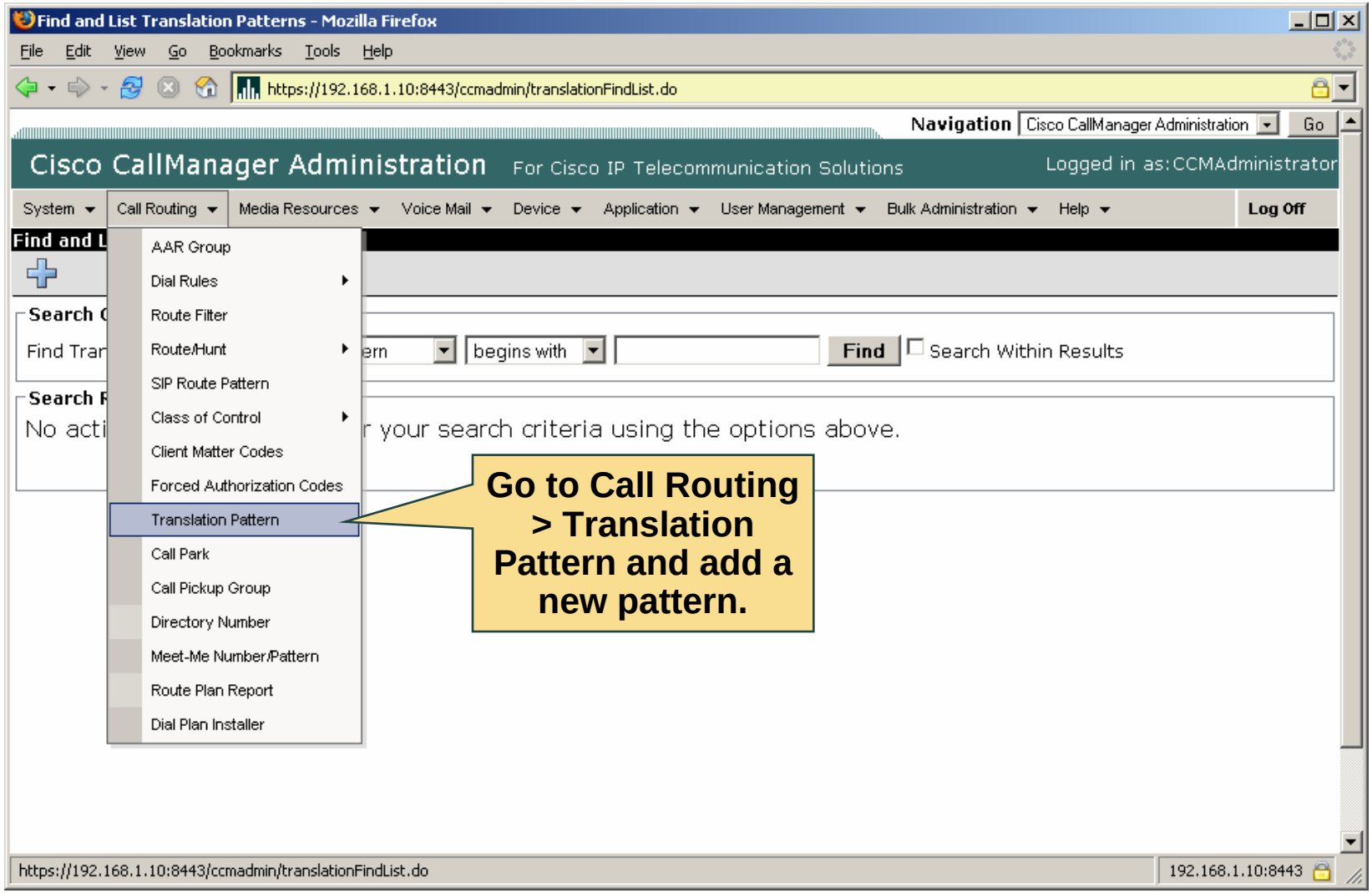
Configuring On-Cluster Intersite Routing



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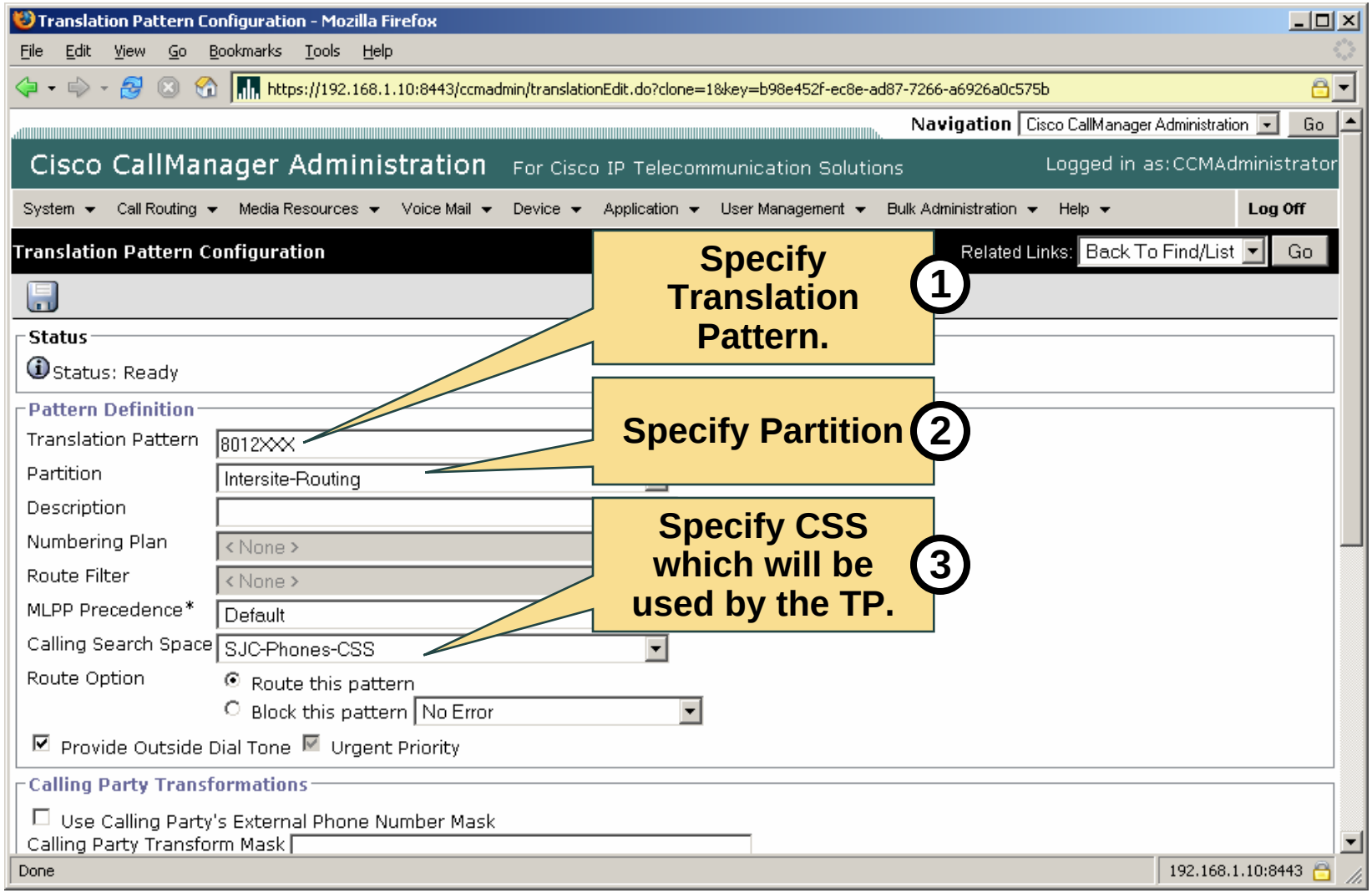
Configuring On-Cluster Intersite Routing



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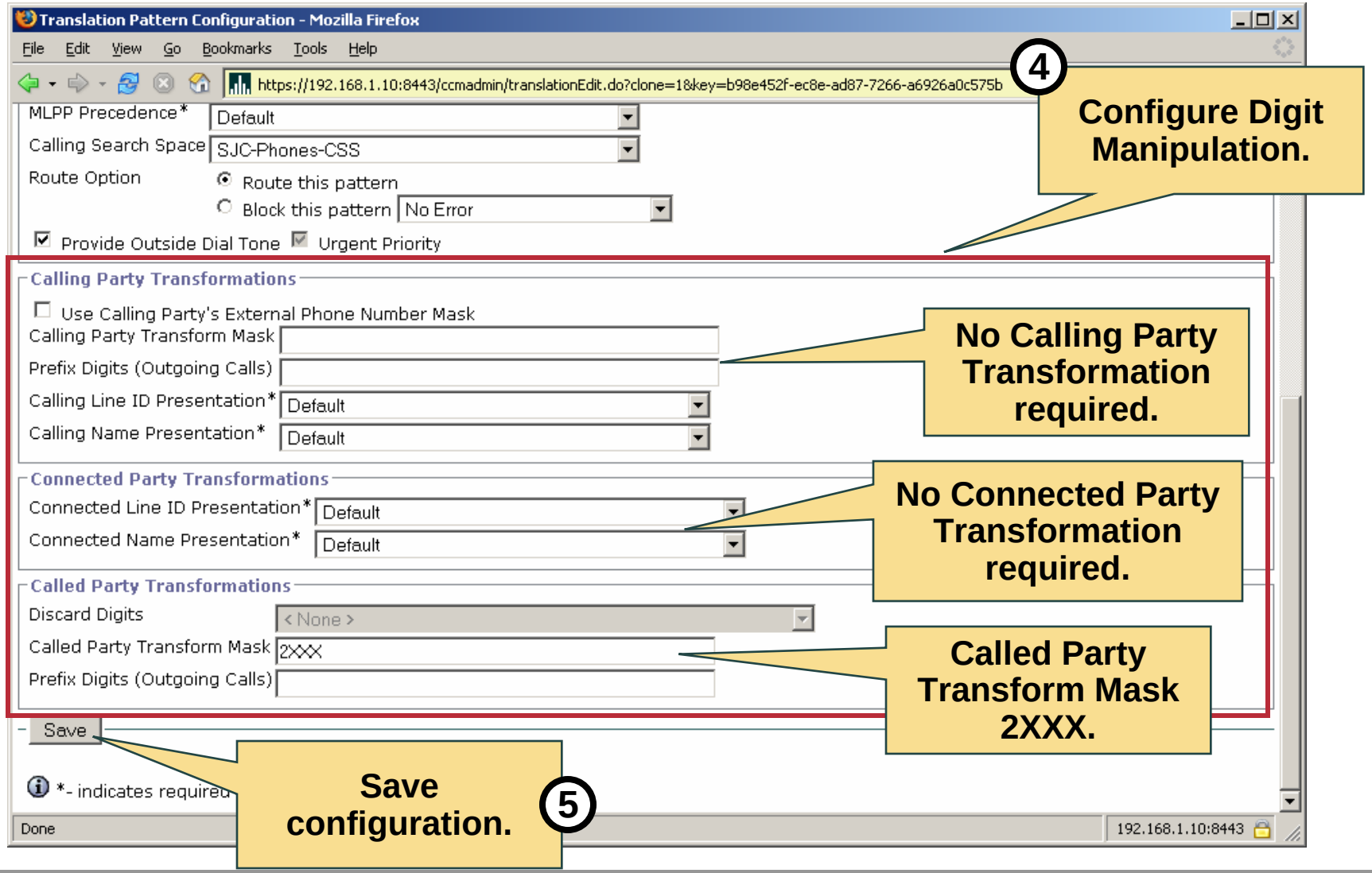
Configuring On-Cluster Intersite Routing



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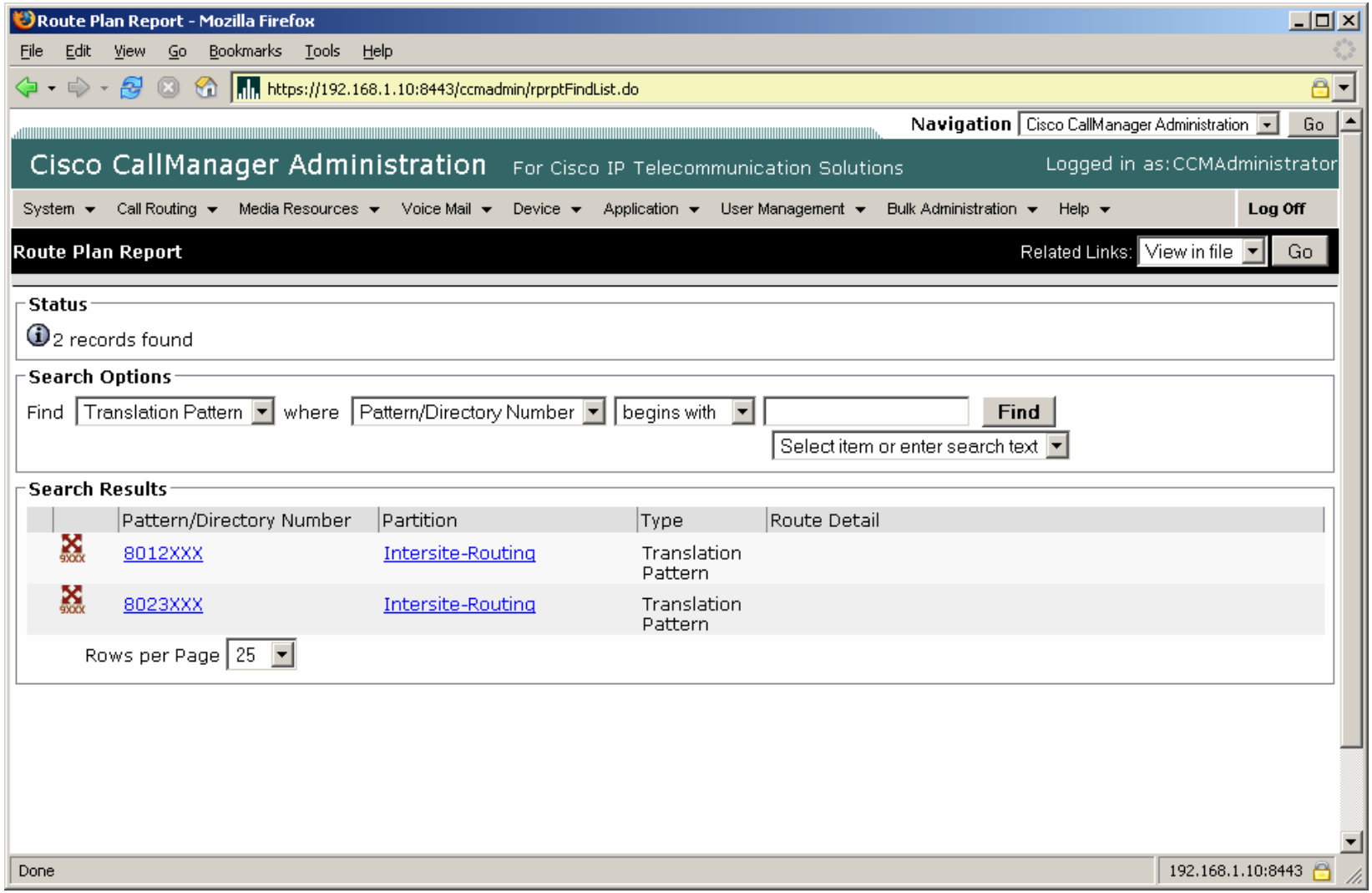
Configuring On-Cluster Intersite Routing



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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.**

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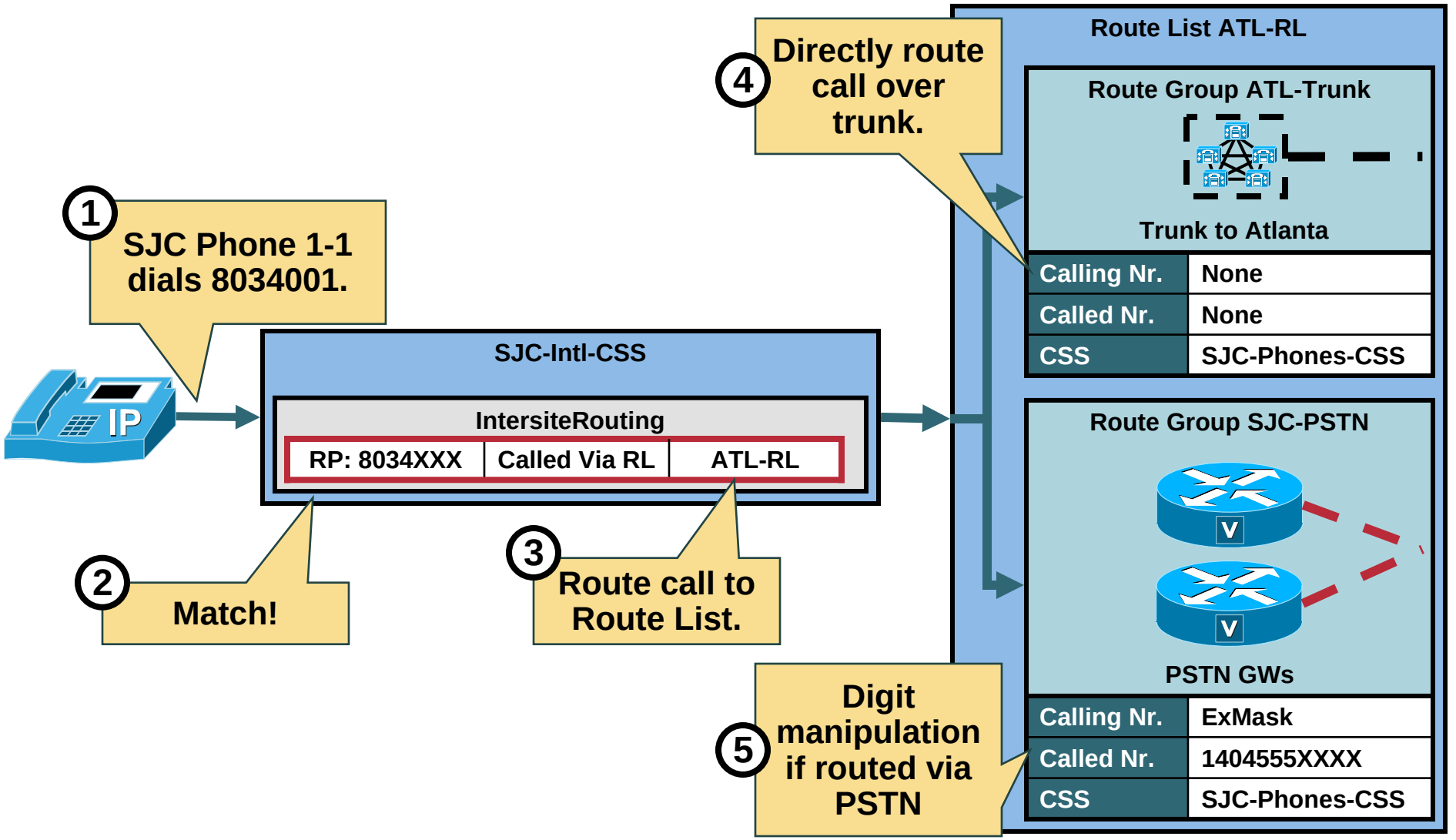
Configuring Off-Cluster Intersite Routing



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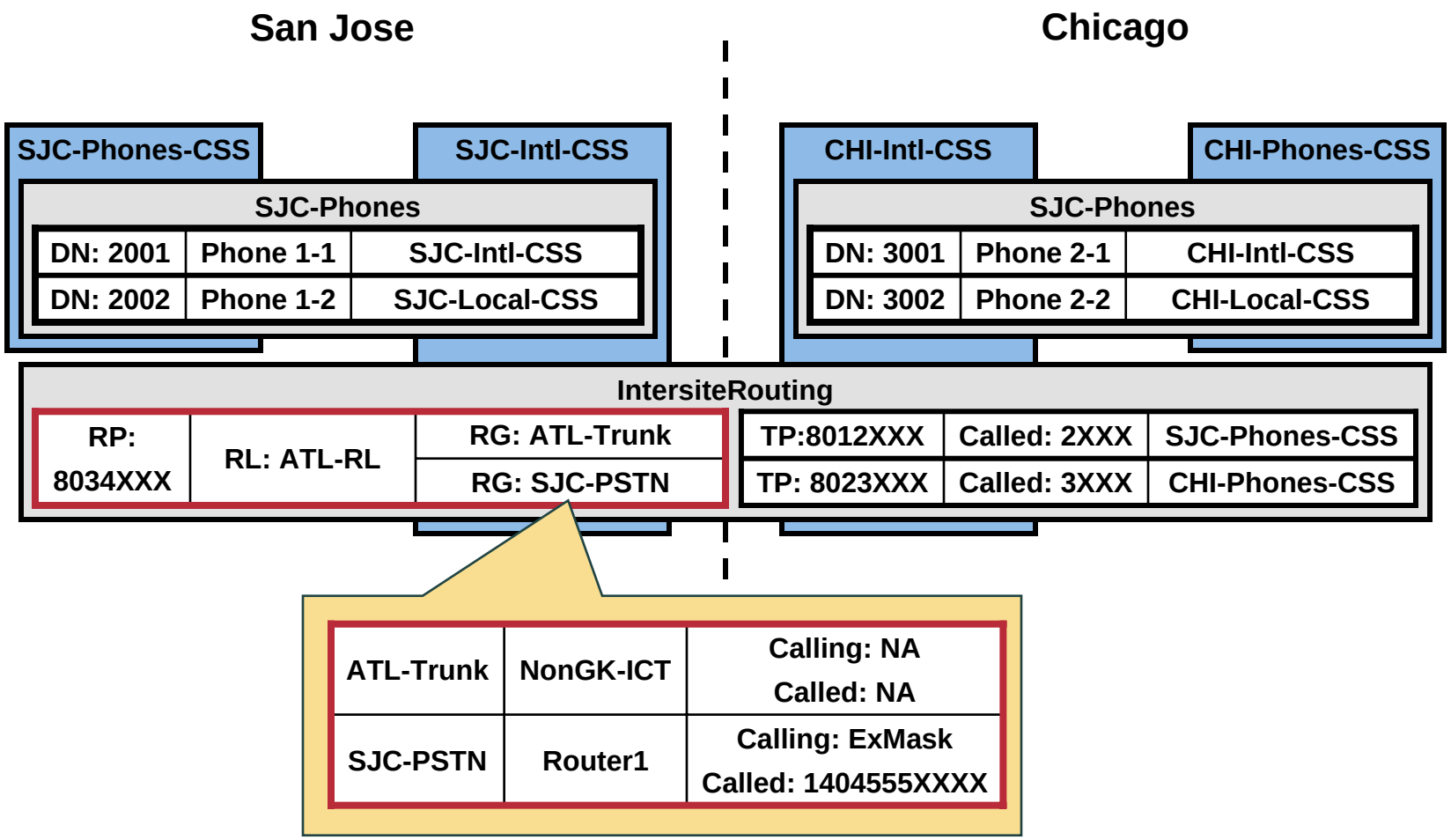
Configuring Off-Cluster Intersite Routing



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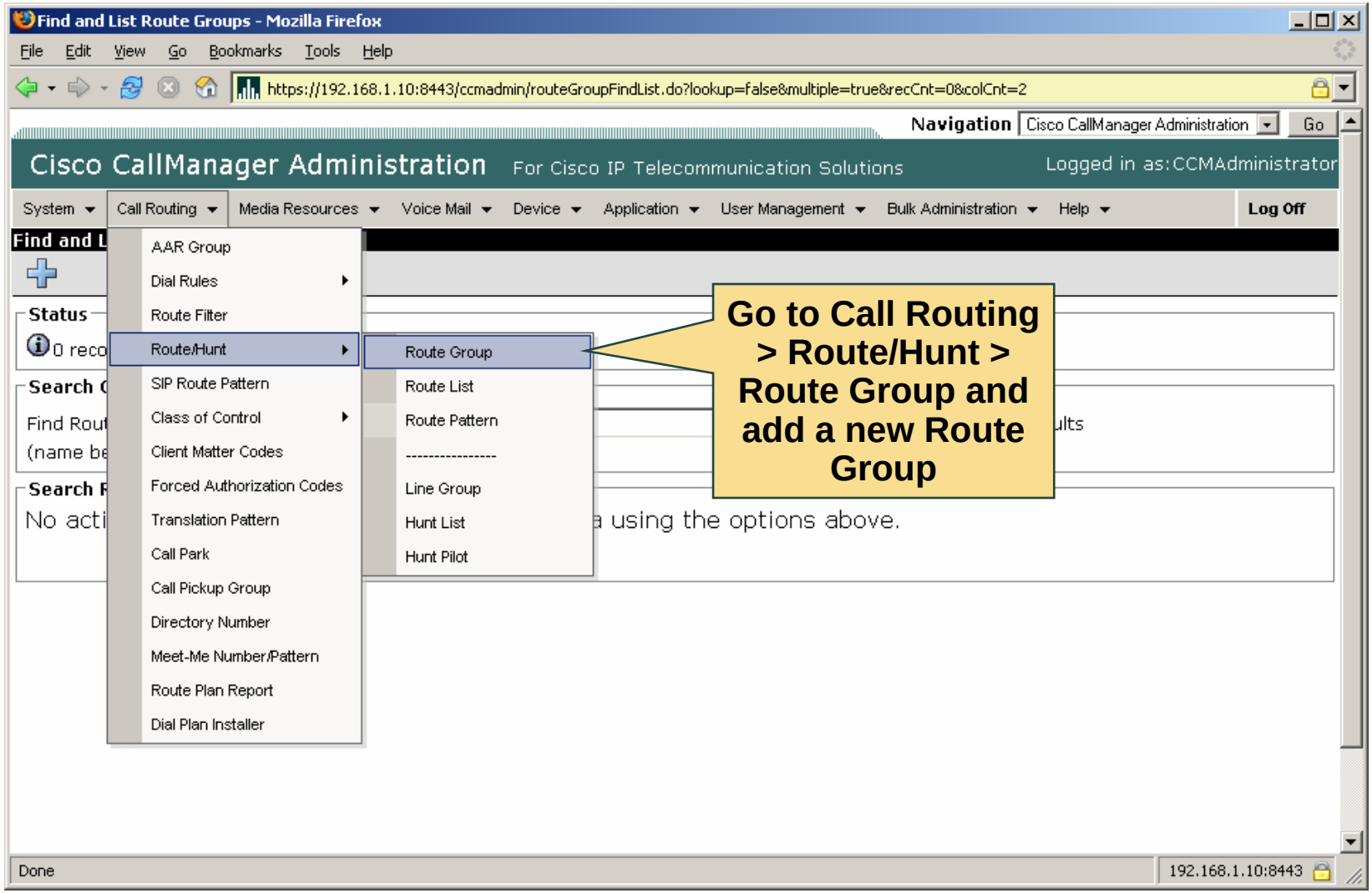
Configuring Off-Cluster Intersite Routing: Adding Route Groups



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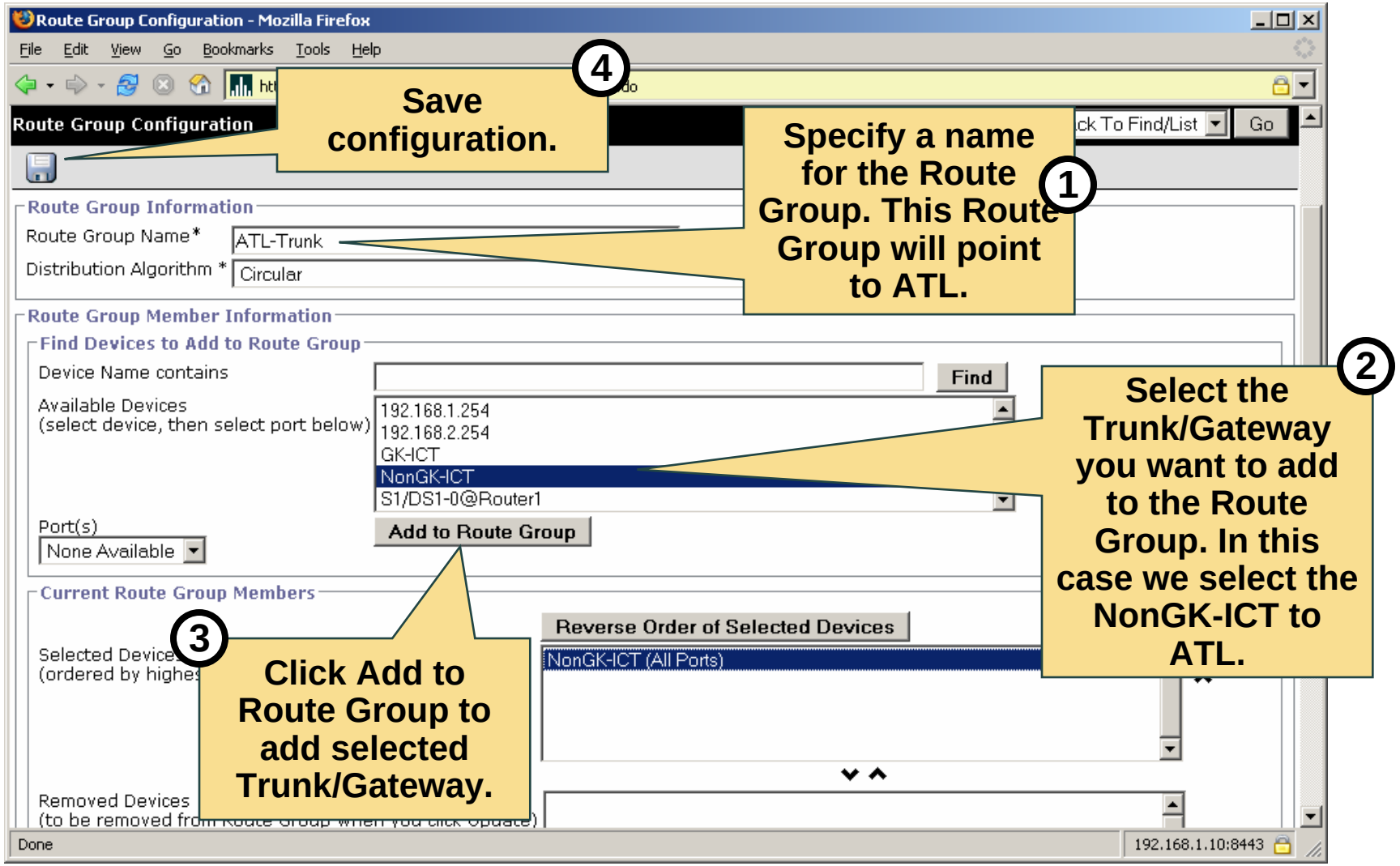
Configuring Off-Cluster Intersite Routing: Configuring Route Groups



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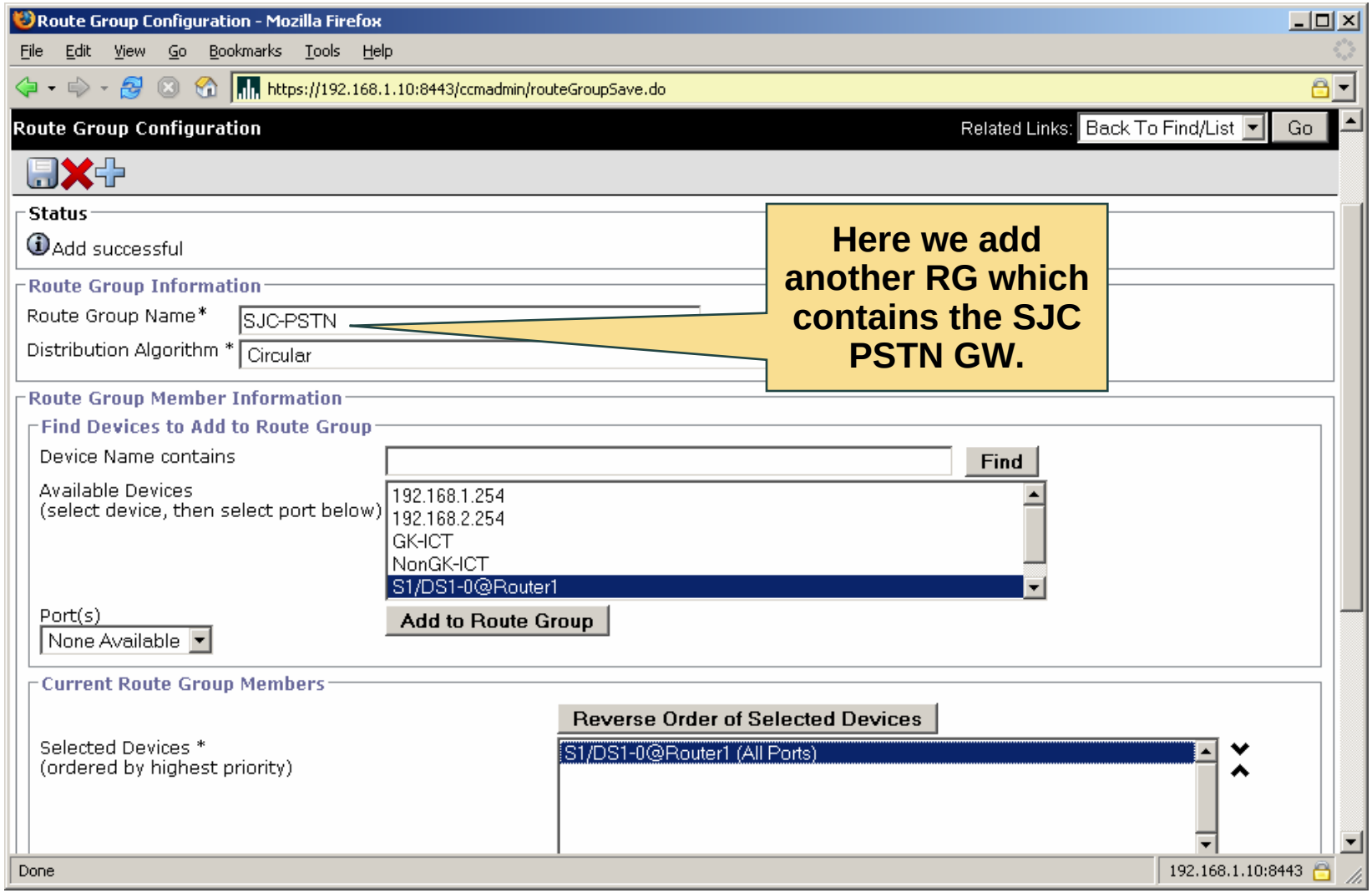
Configuring Off-Cluster Intersite Routing: Configuring Route Groups



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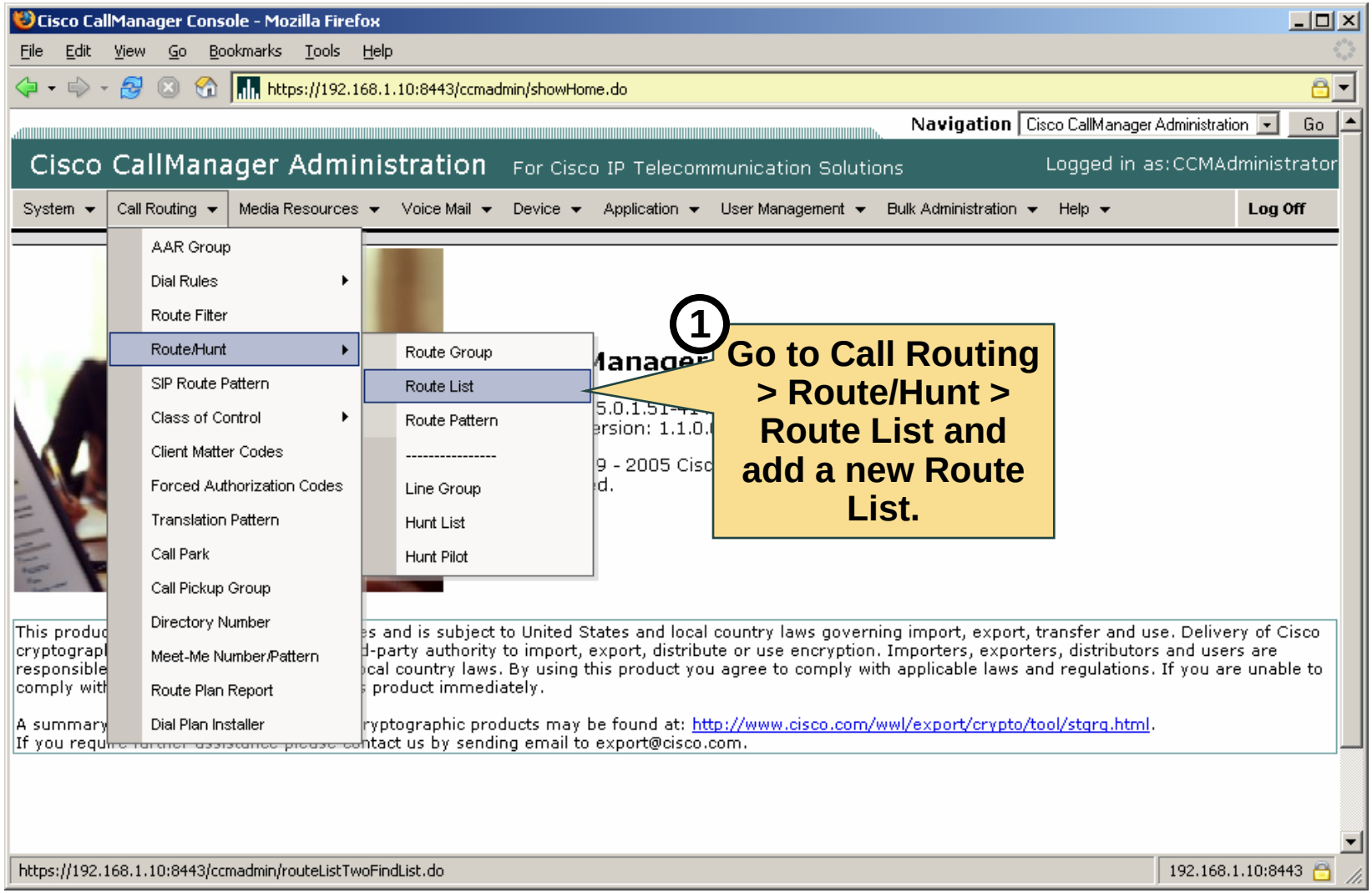
Configuring Off-Cluster Intersite Routing: Adding Route Lists



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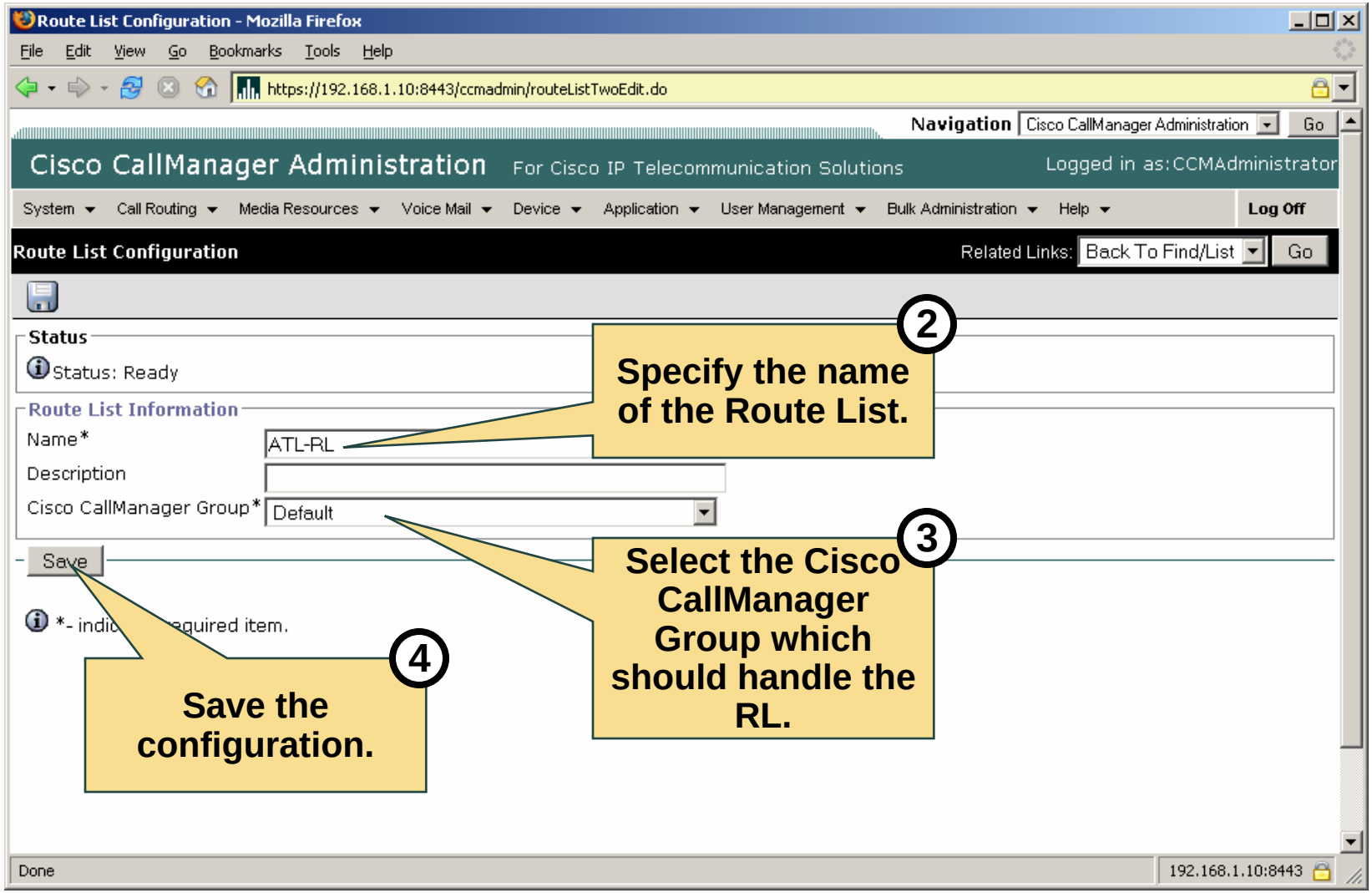
Configuring Off-Cluster Intersite Routing: Adding Route Lists



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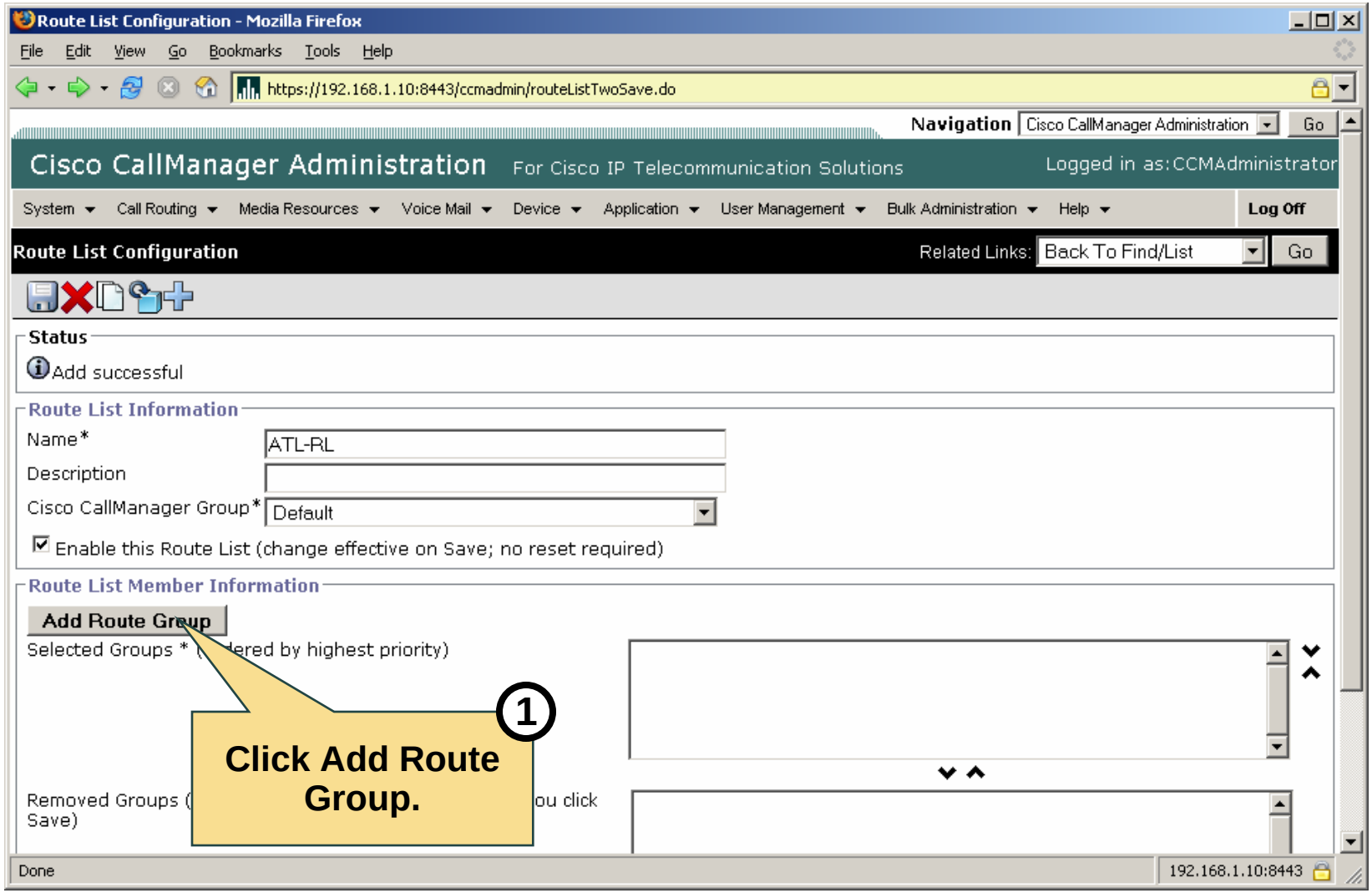
Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists



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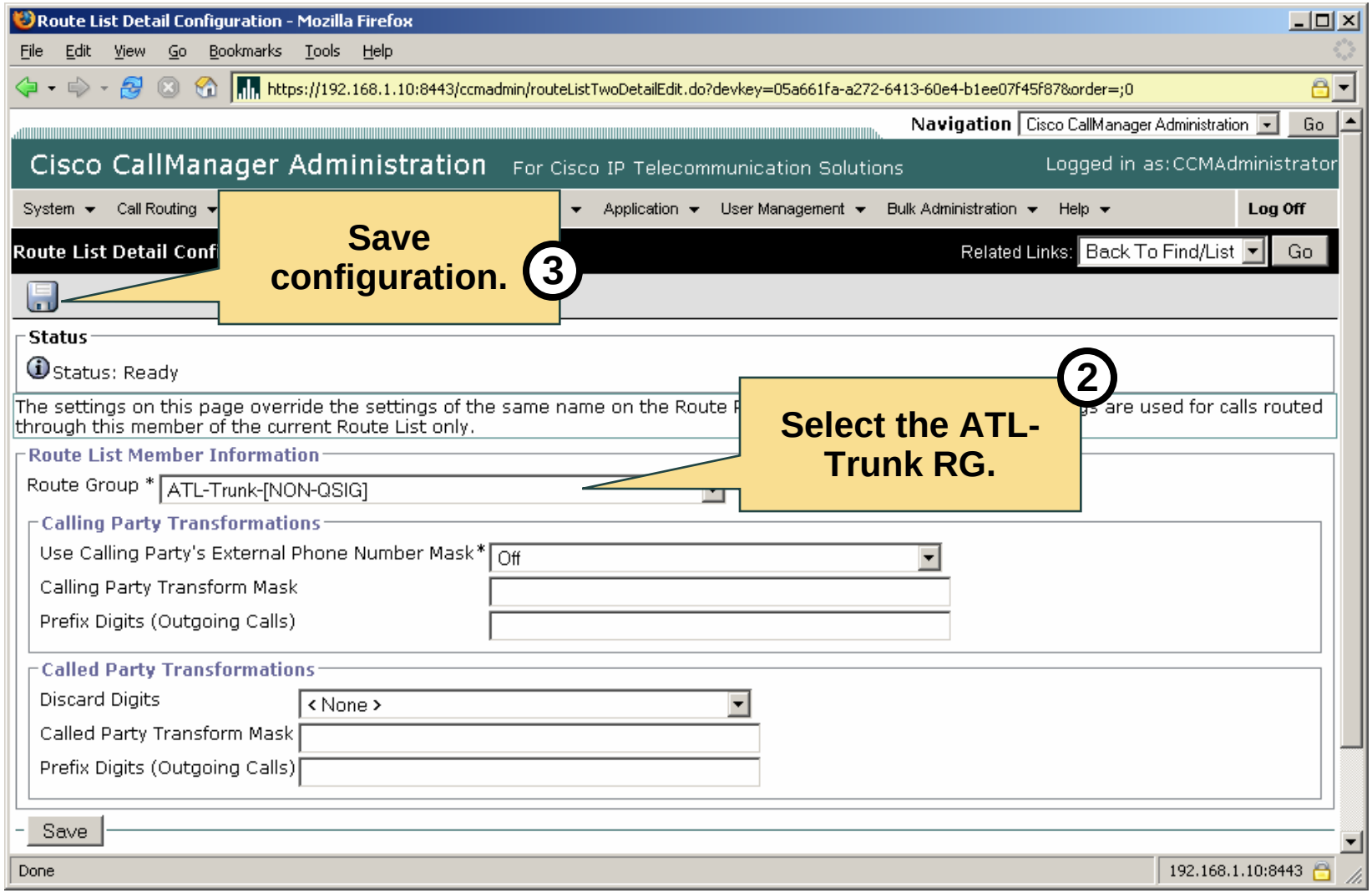
Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists



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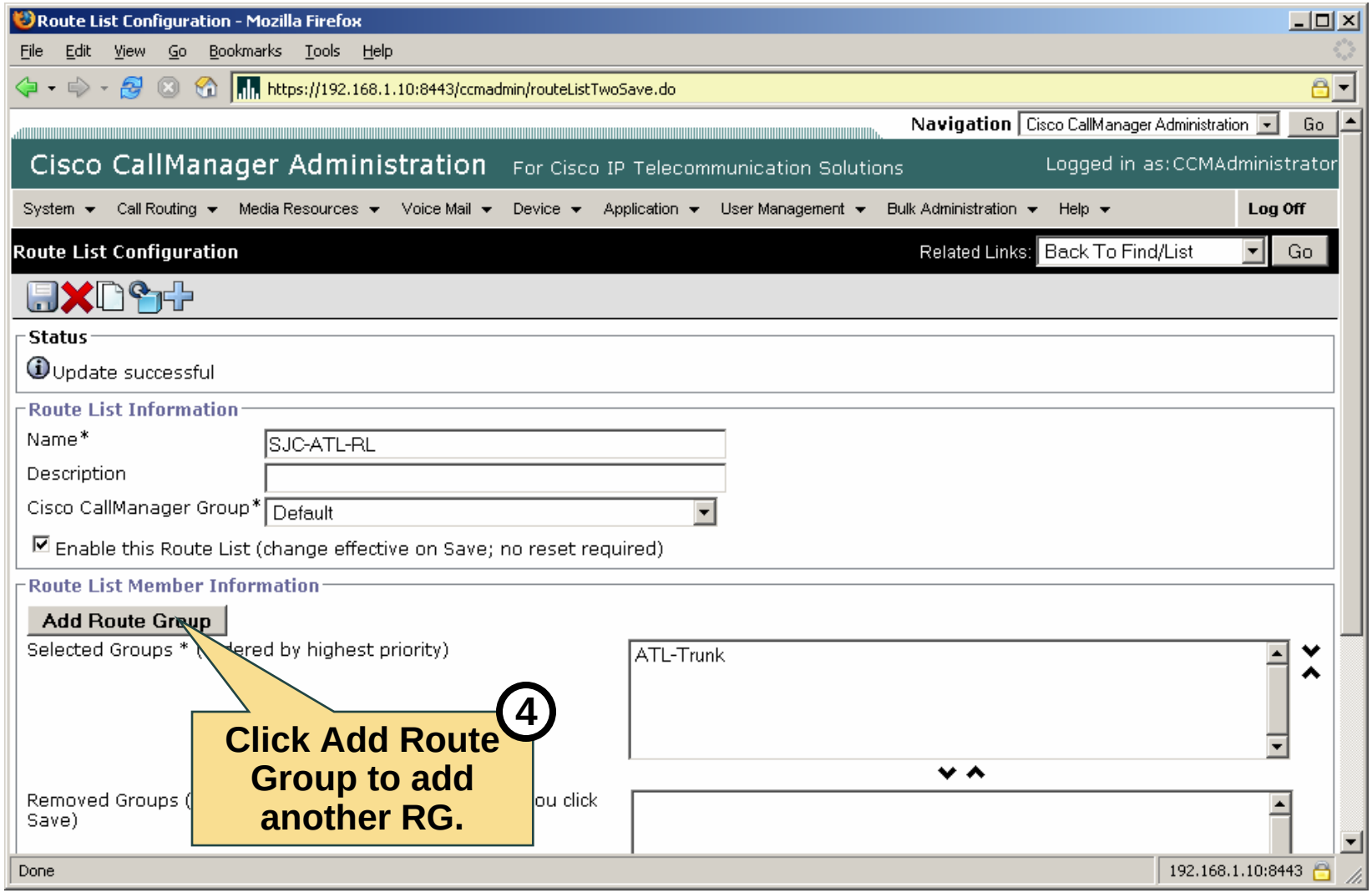
Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists



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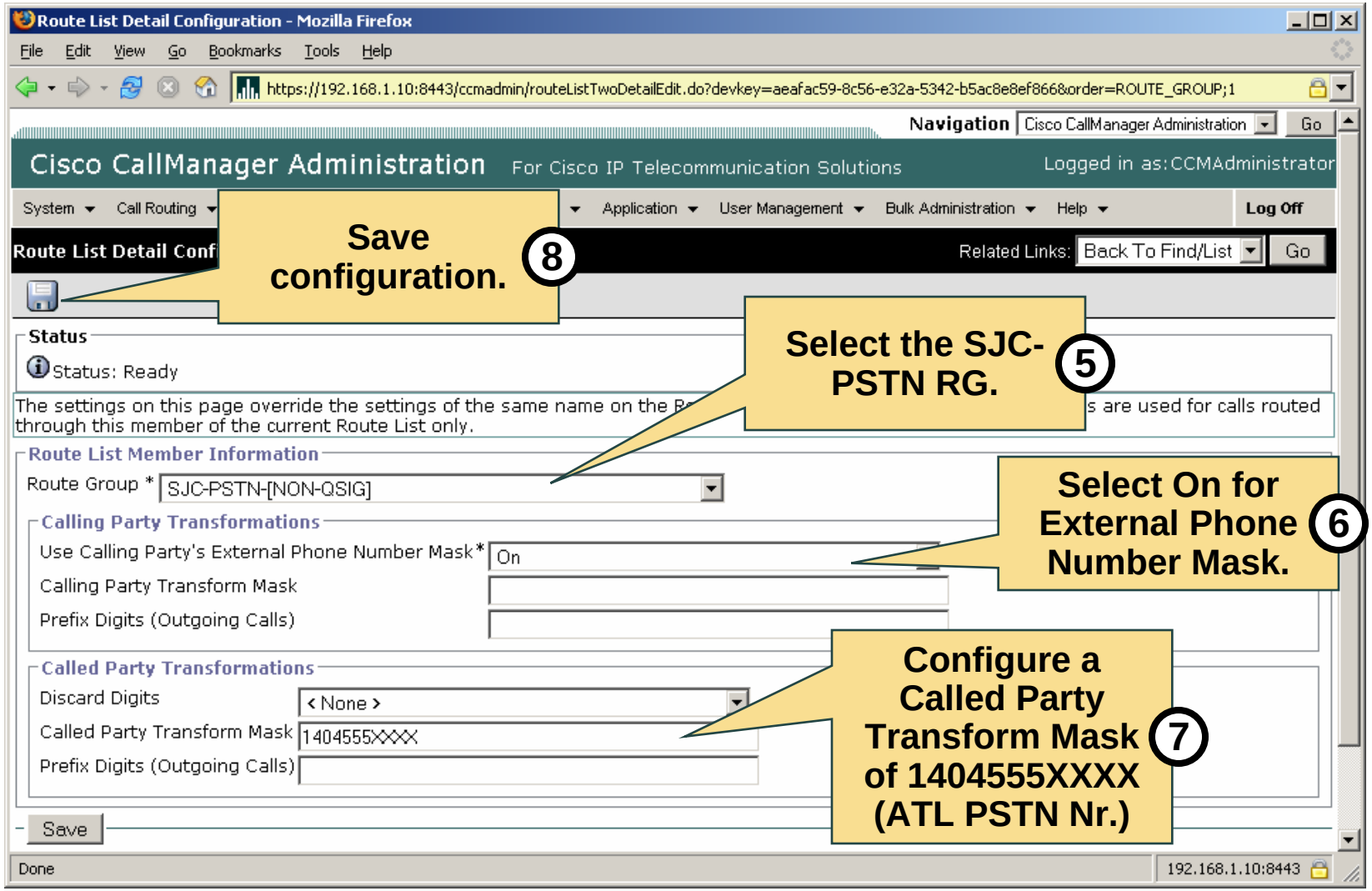
Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists



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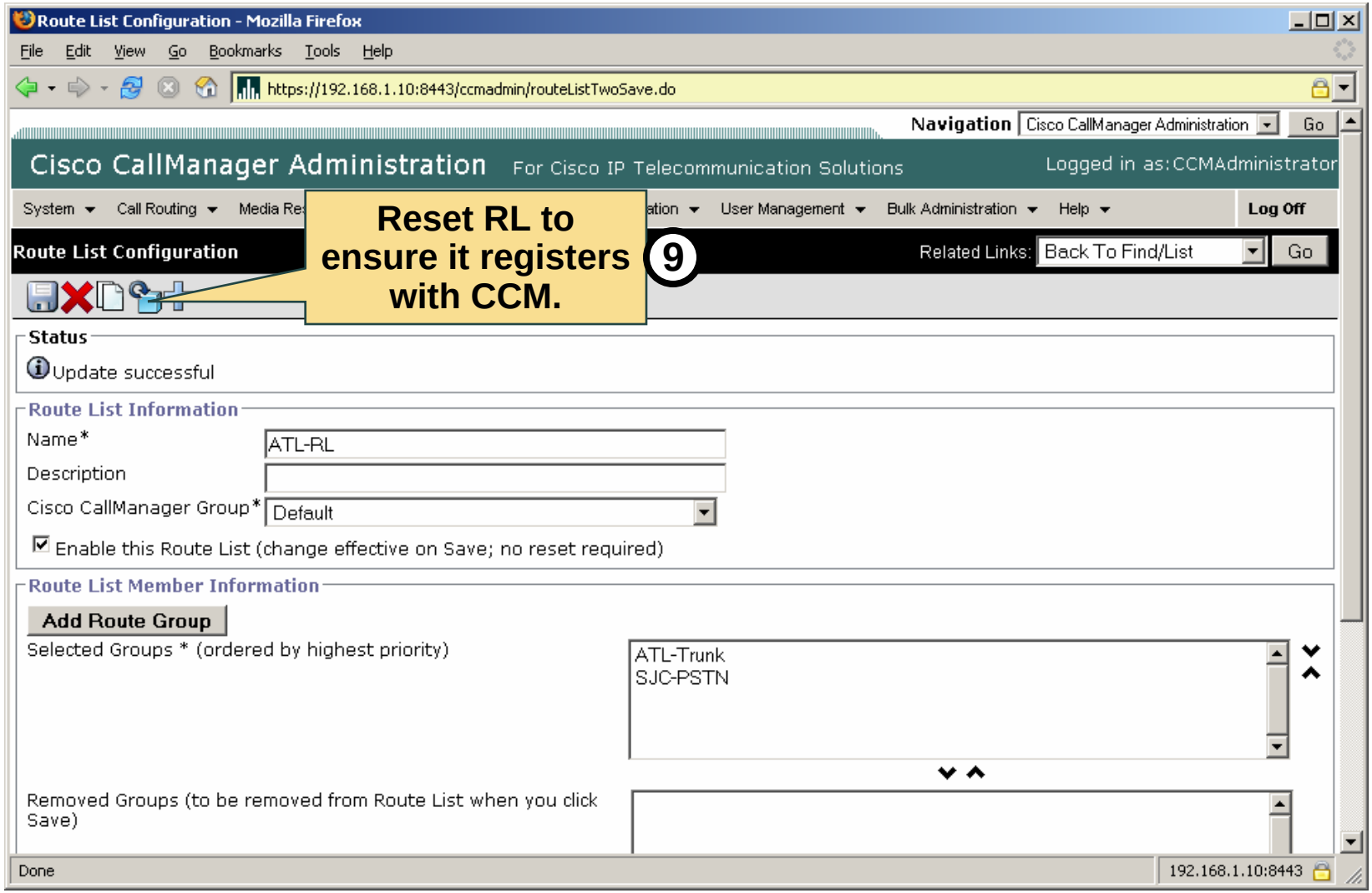
Configuring Off-Cluster Intersite Routing: Adding Route Groups to Route Lists



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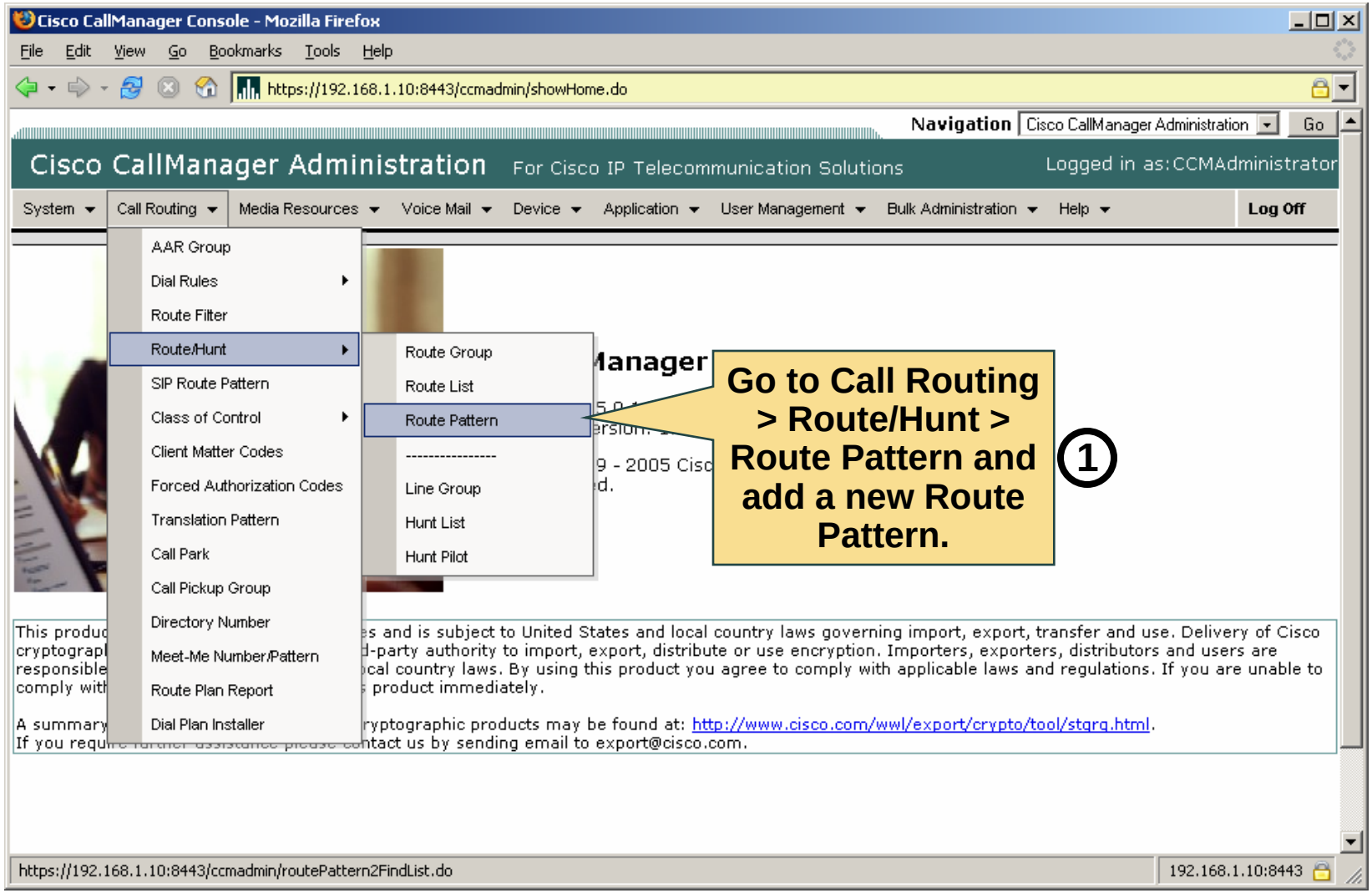
Configuring Off-Cluster Intersite Routing: Configuring Route Patterns



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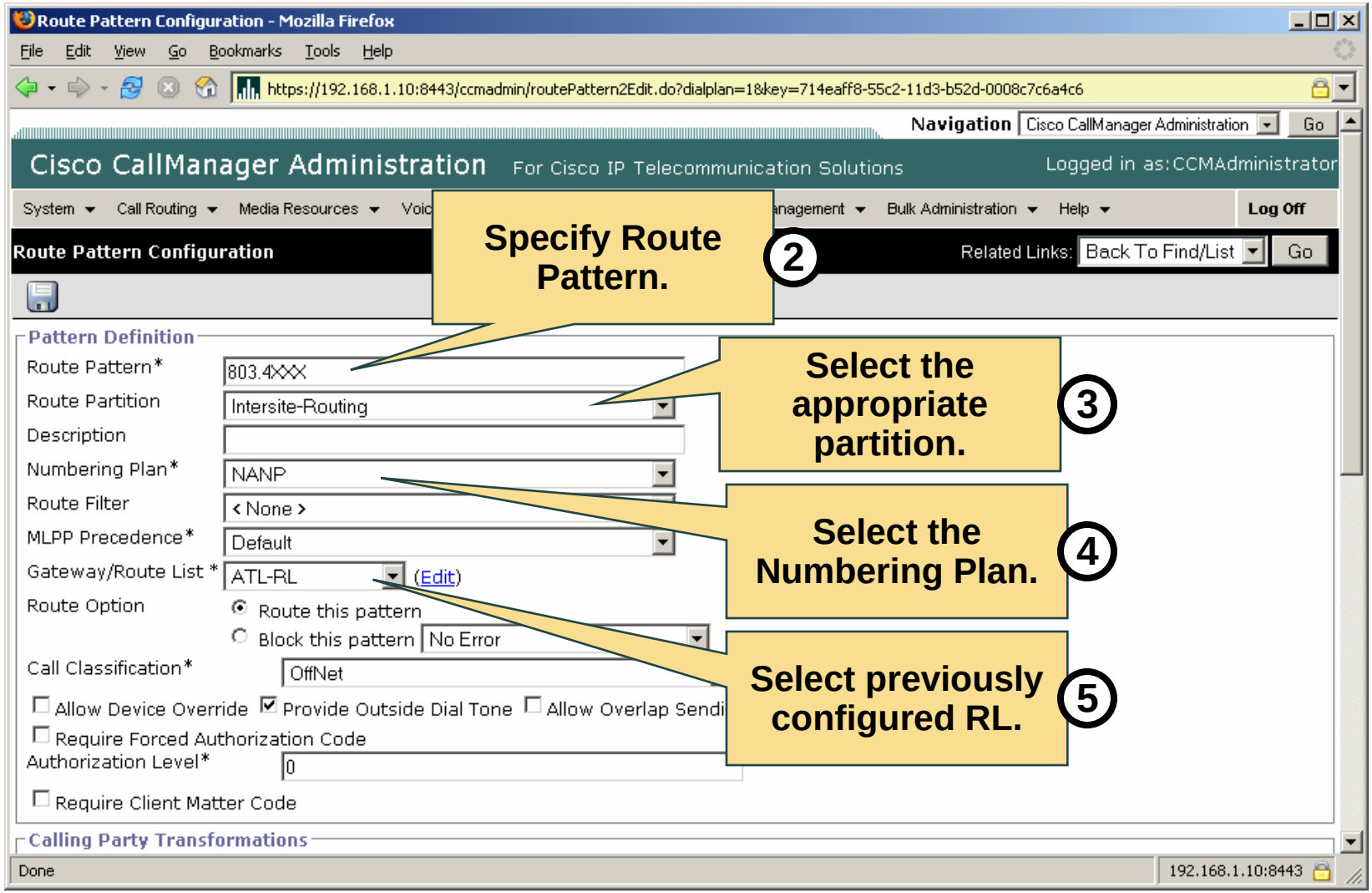
Configuring Off-Cluster Intersite Routing: Configuring Route Patterns



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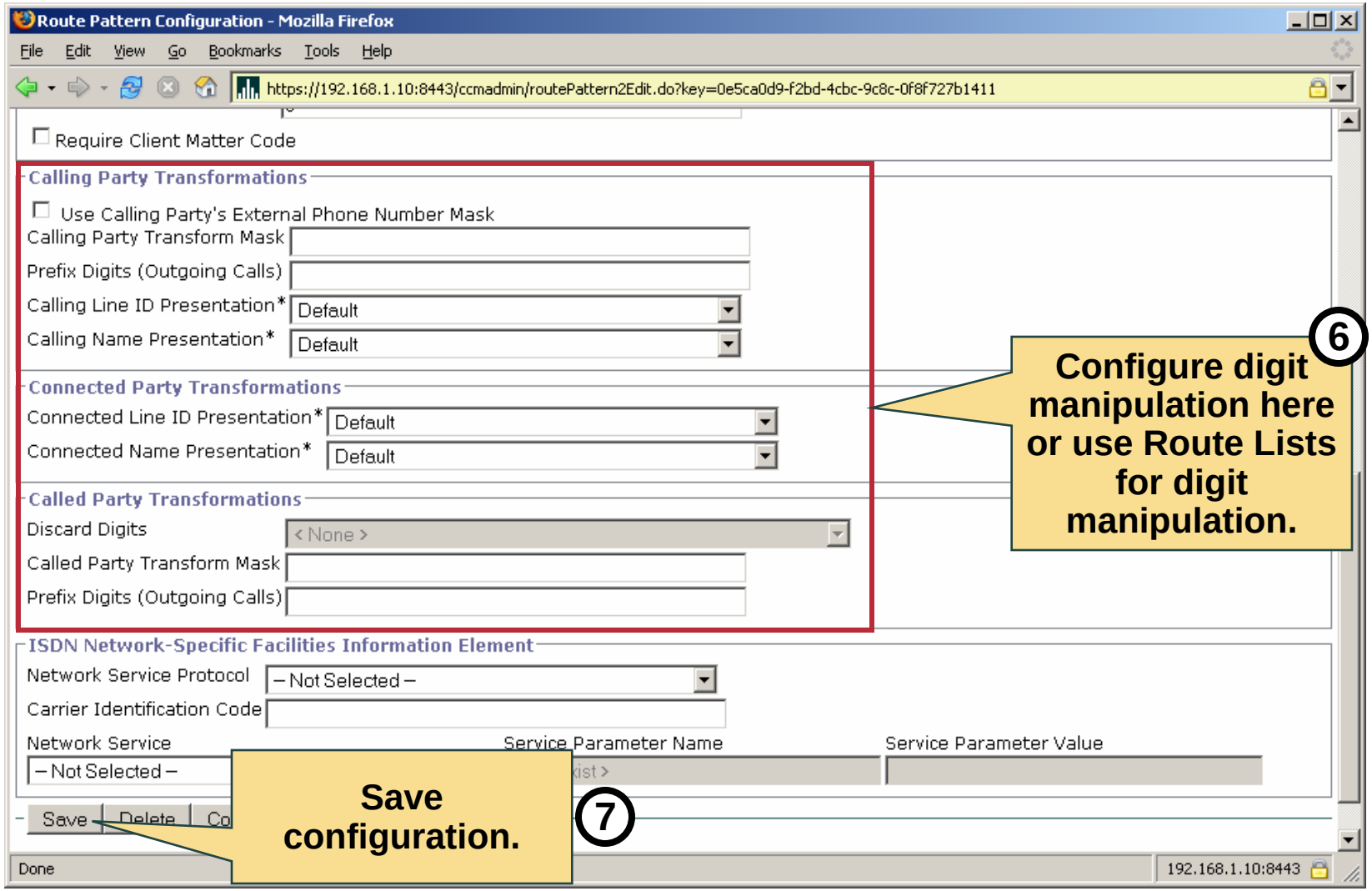
Configuring Off-Cluster Intersite Routing: Configuring Route Patterns



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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.**

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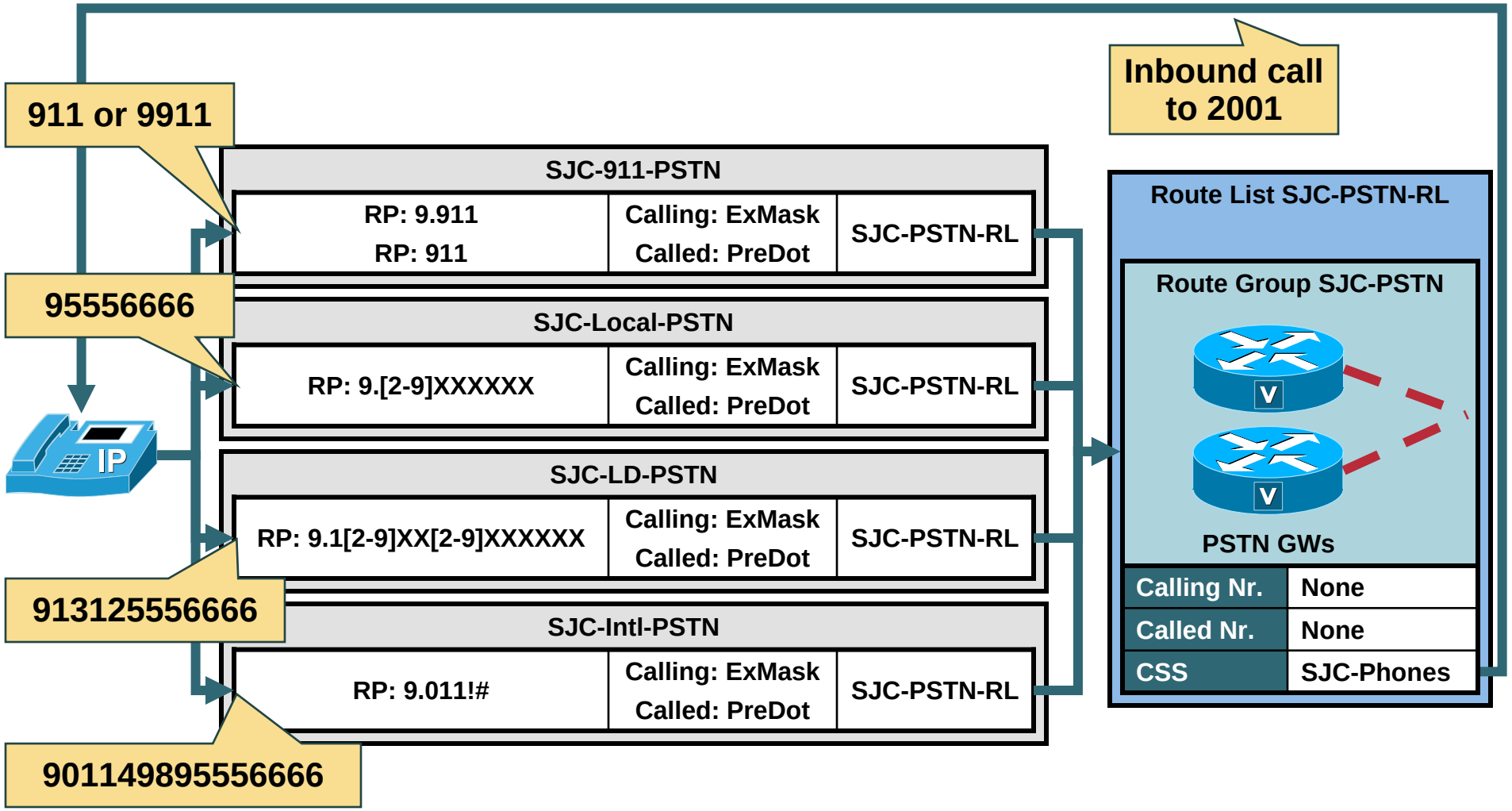
Configuring PSTN Routing and Calling Privileges



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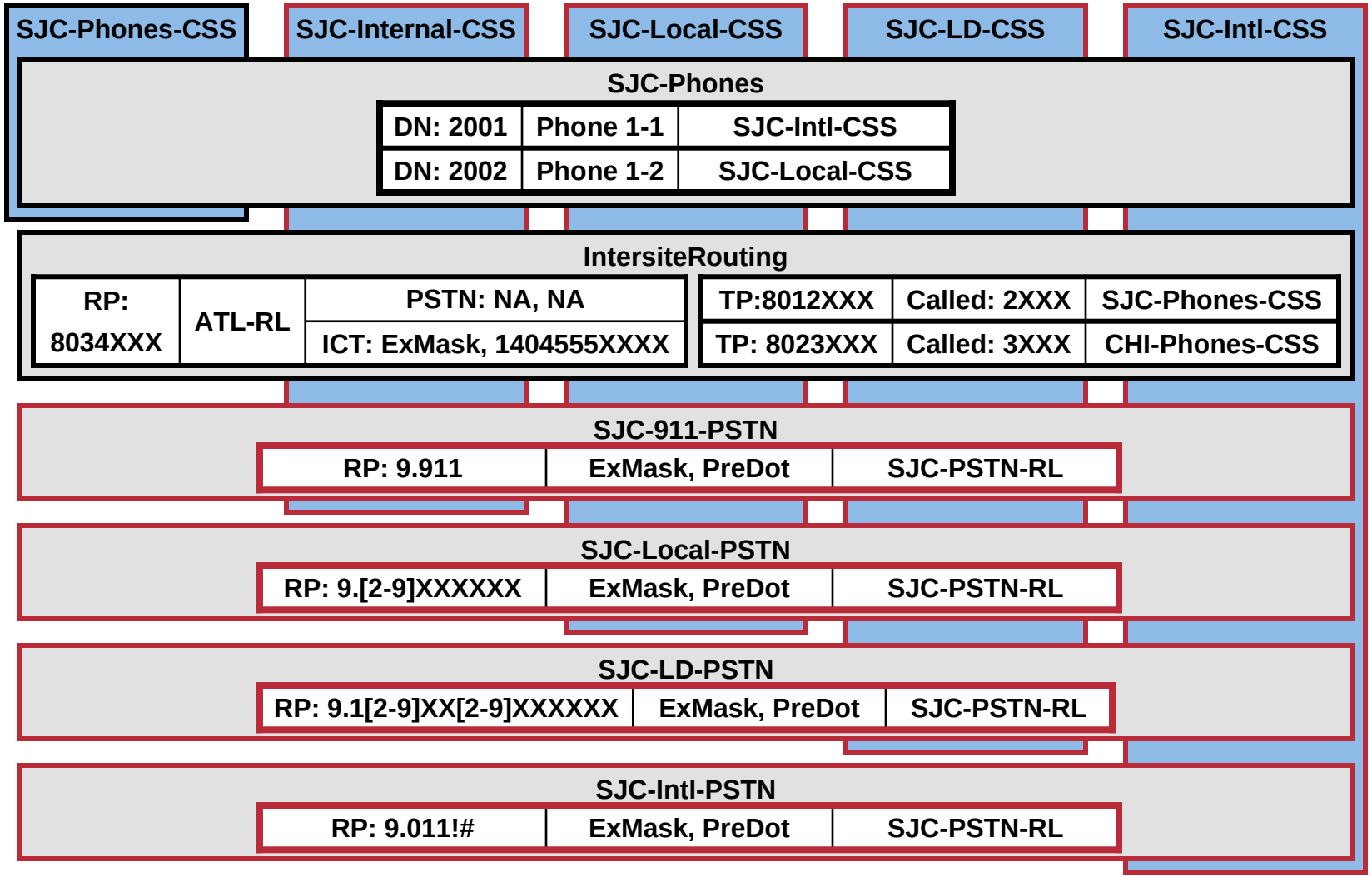
Configuring PSTN Routing and Calling Privileges: San Jose



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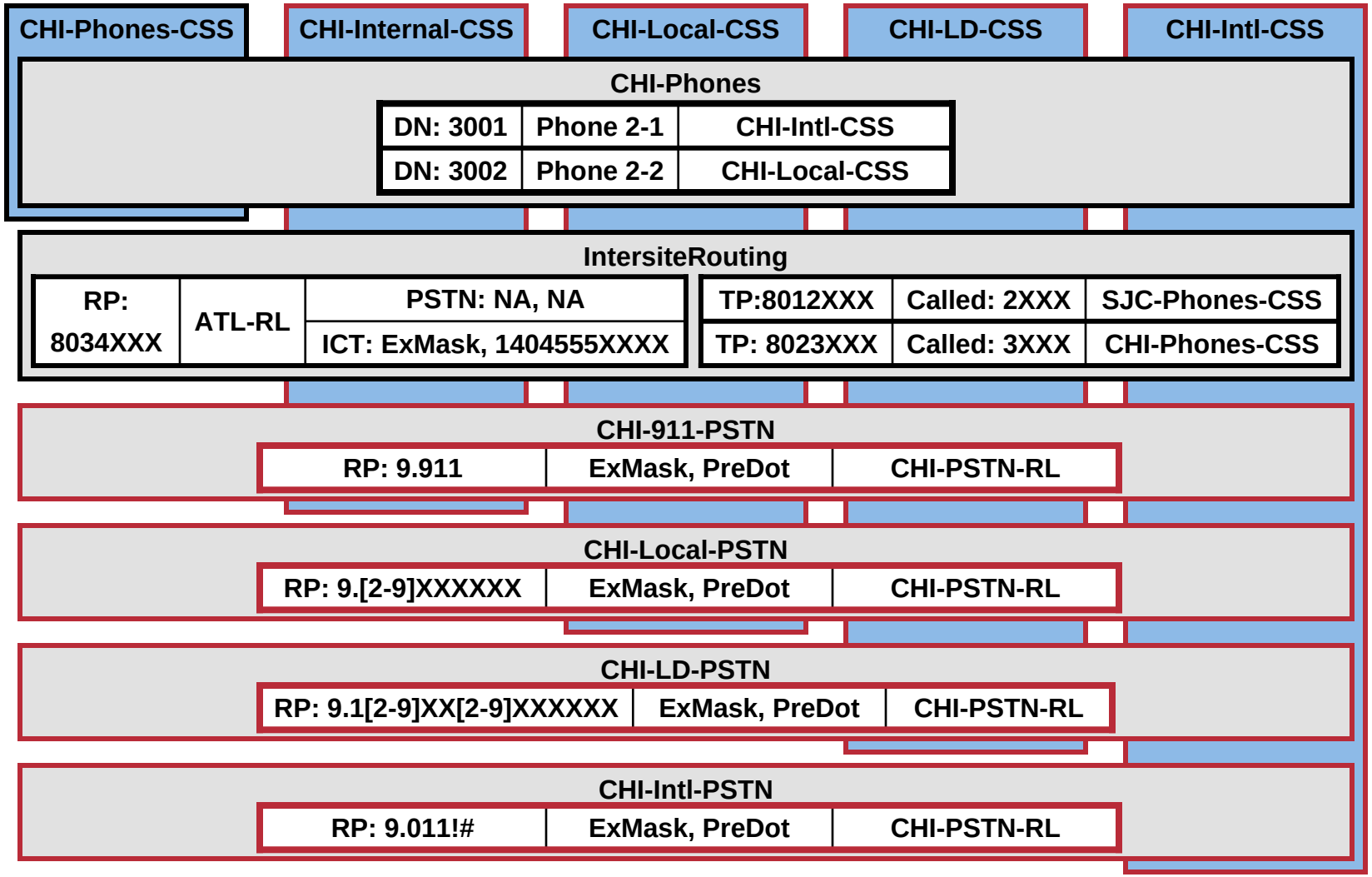
Configuring PSTN Routing and Calling Privileges: Chicago



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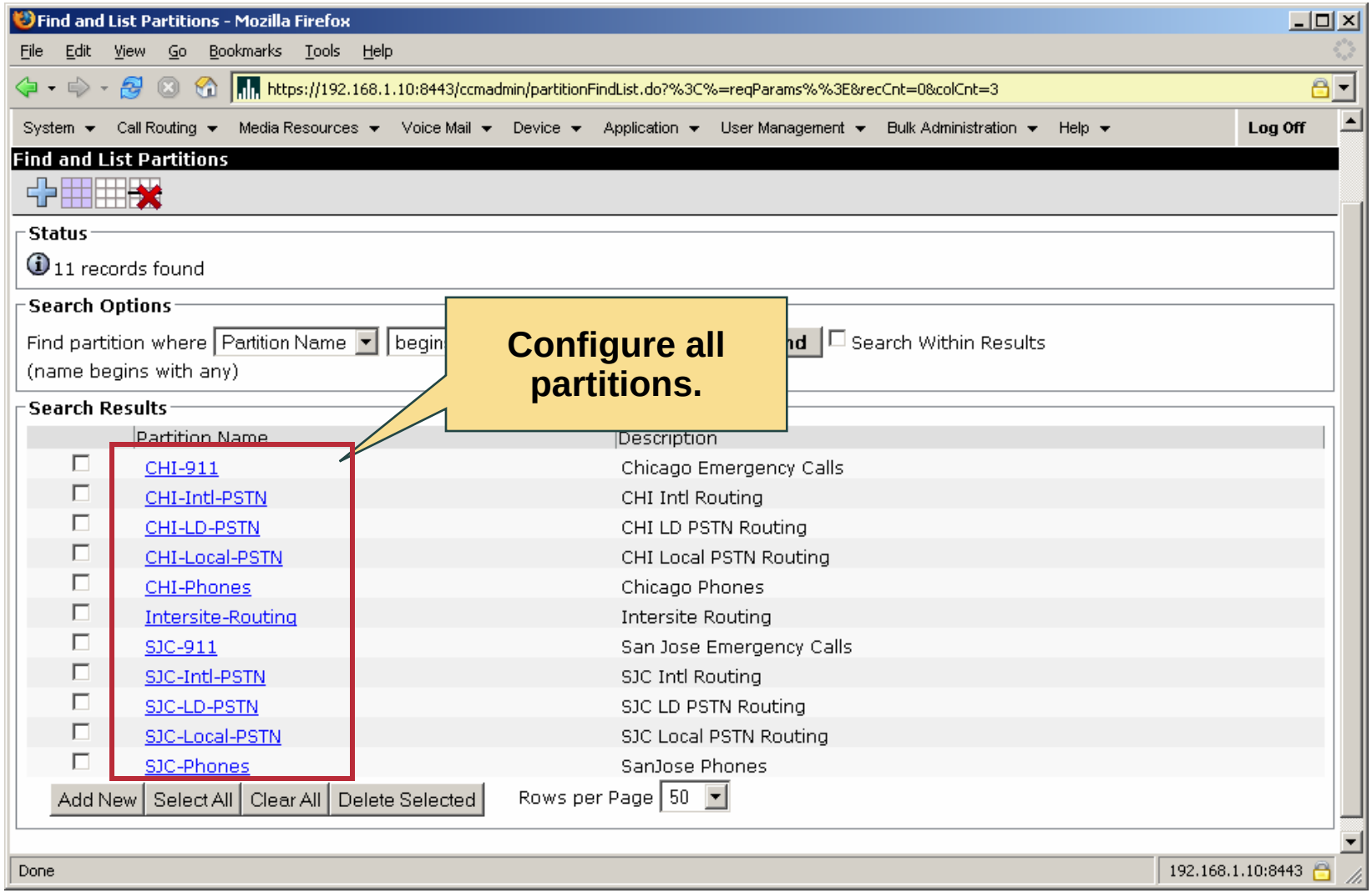
Configuring PSTN Routing and Calling Privileges: Revised Partitions and CSSs



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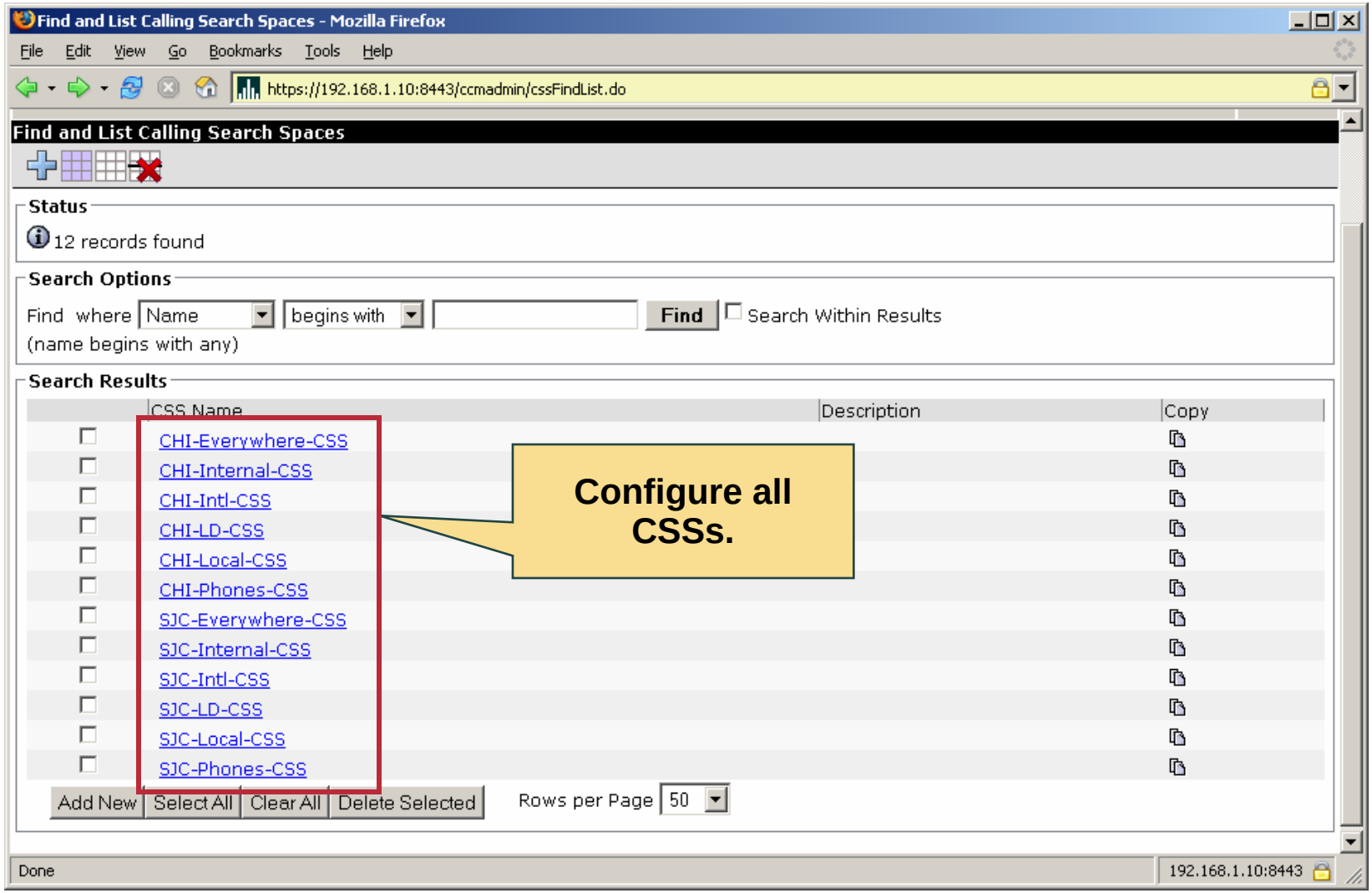
Configuring PSTN Routing and Calling Privileges: Revised Partitions and CSSs



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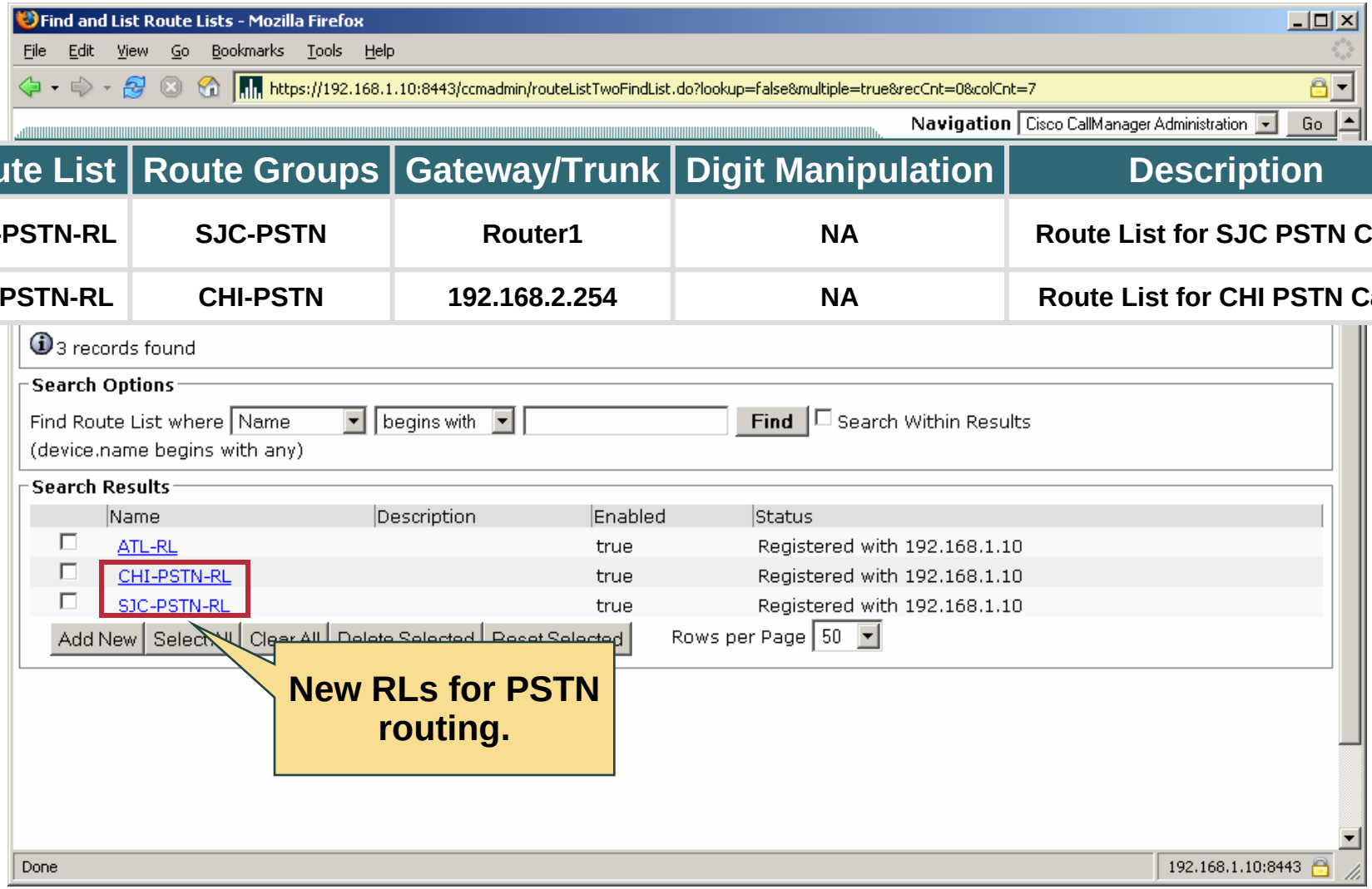
Configuring PSTN Routing and Calling Privileges: Additional Route Lists

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



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Configuring PSTN Routing and Calling Privileges: Additional Route Patterns

Pattern	Partition	Type	Destination	Digit Manipulation	Description
PSTN Routing					
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

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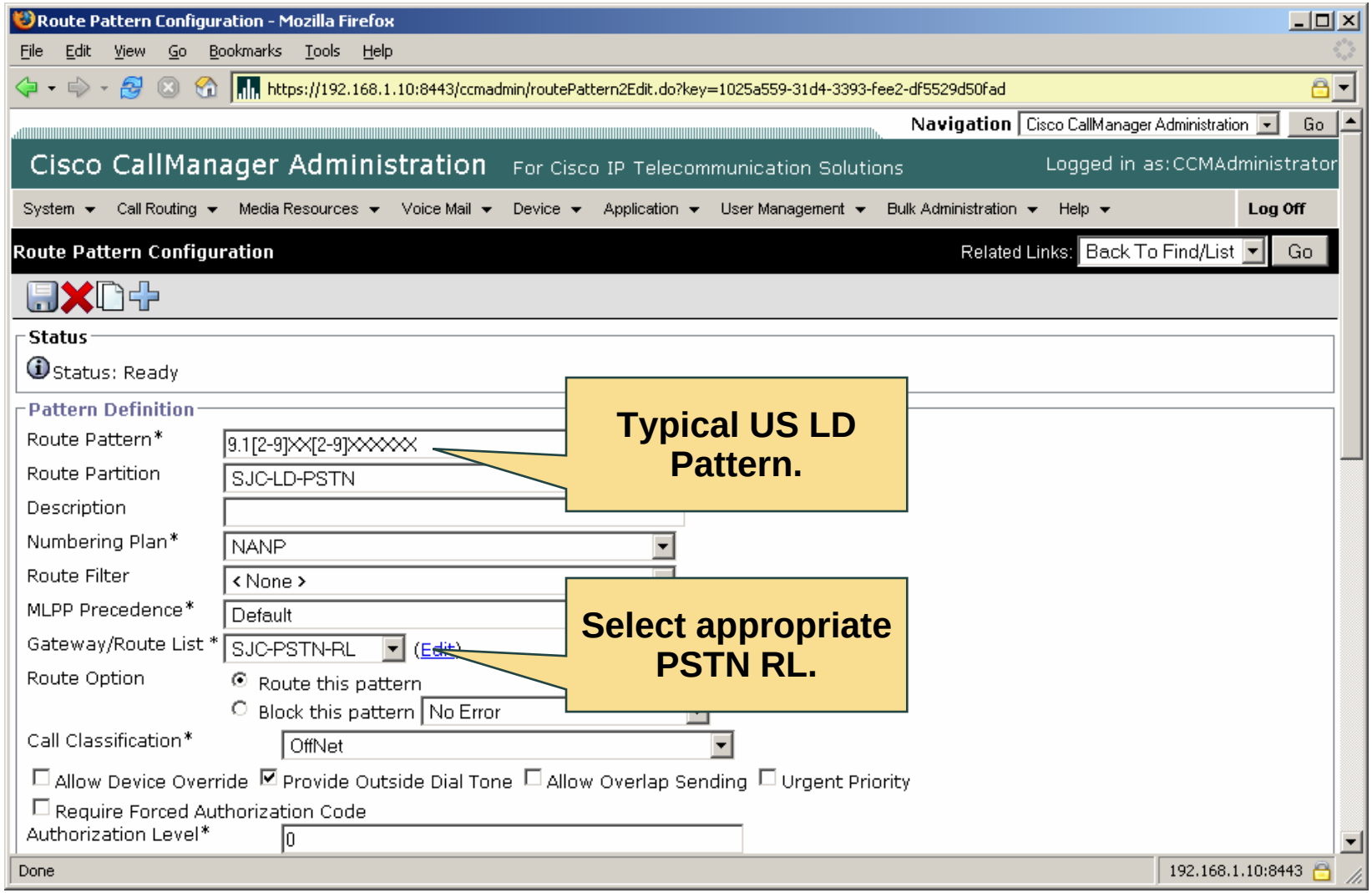
Configuring PSTN Routing and Calling Privileges: Additional Route Patterns



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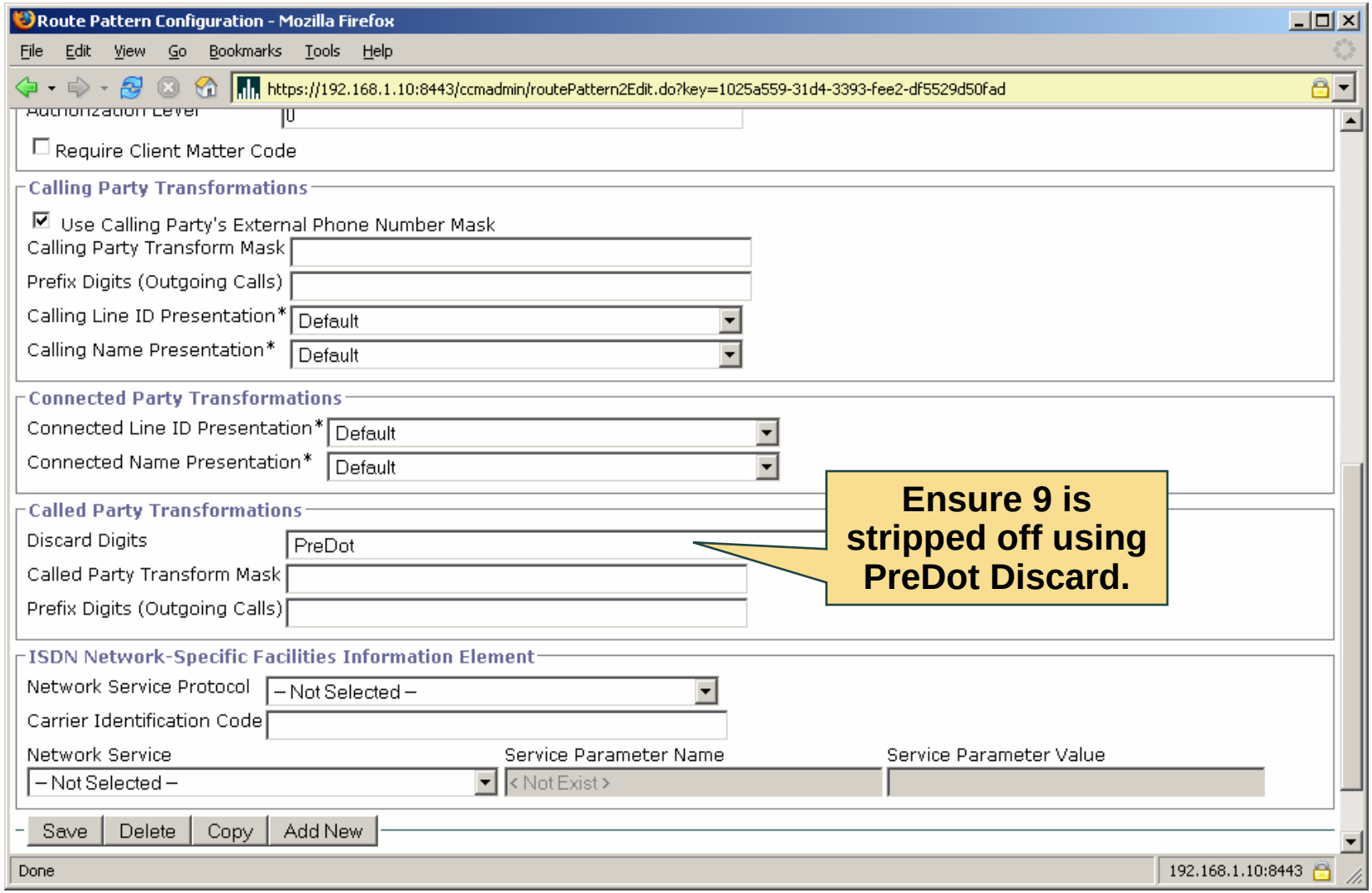
Configuring PSTN Routing and Calling Privileges: Additional Route Patterns



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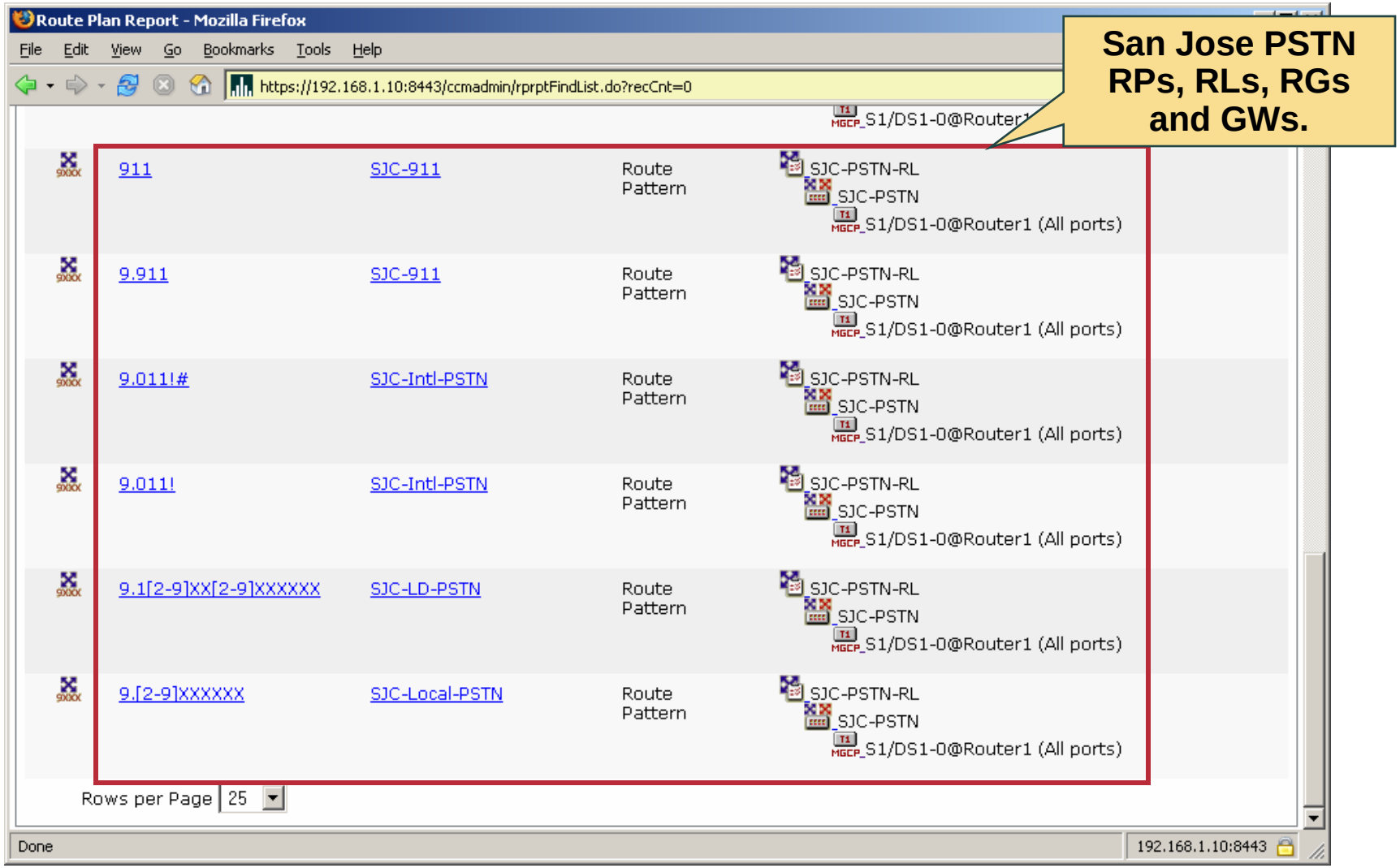
Configuring PSTN Routing and Calling Privileges: Additional Route Patterns



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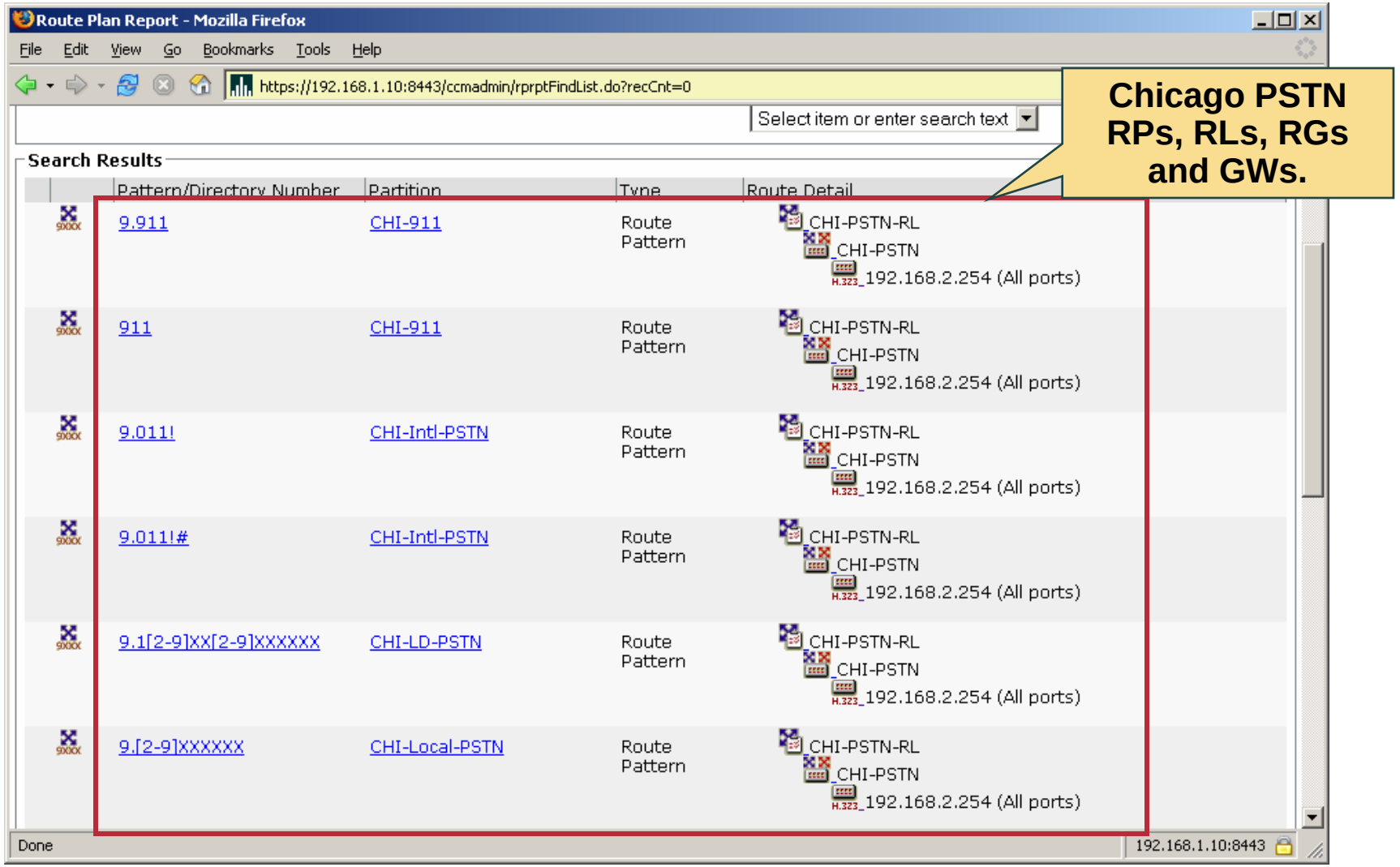
Configuring PSTN Routing and Calling Privileges: Additional Route Patterns



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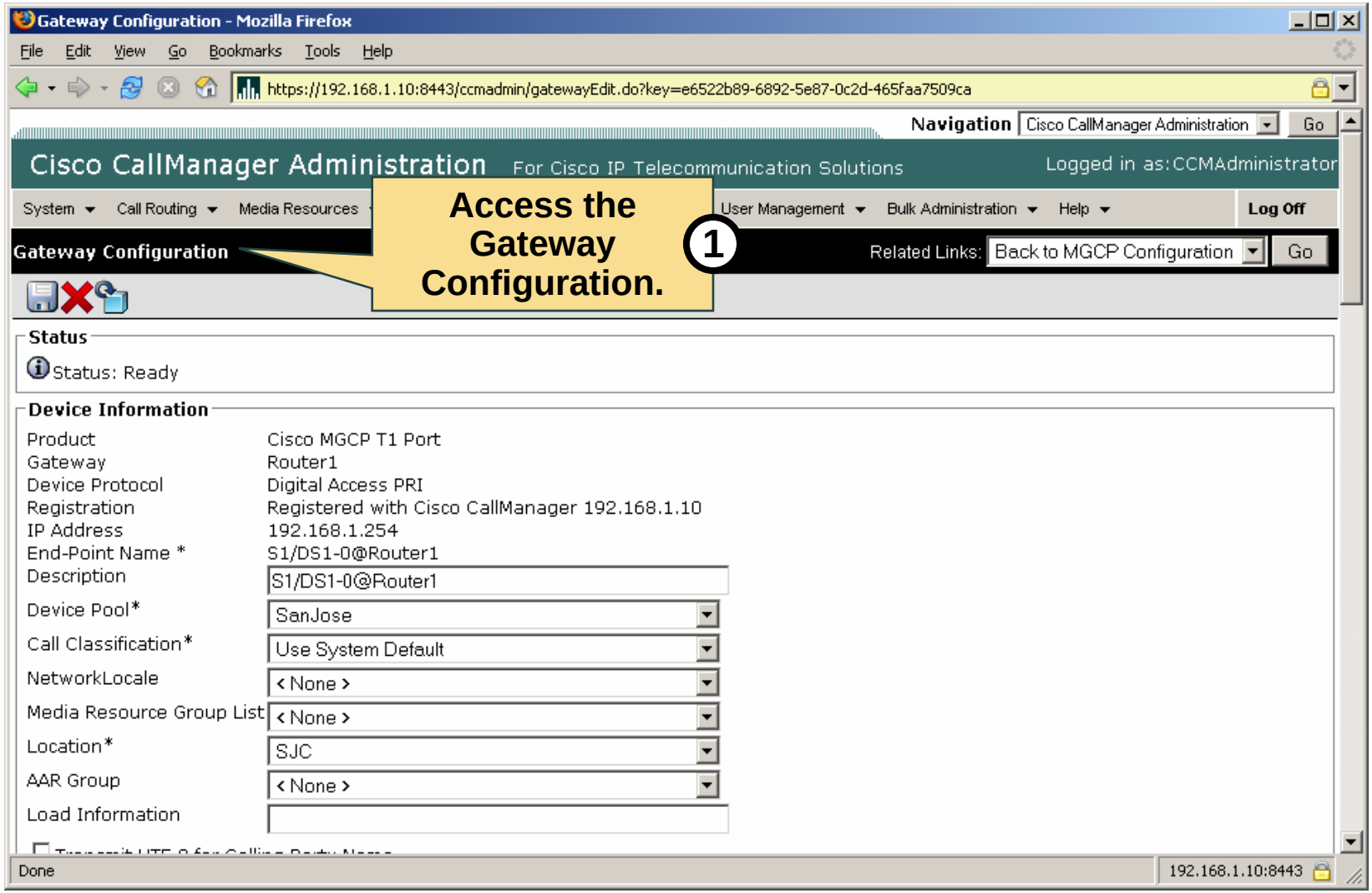
Configuring PSTN Routing and Calling Privileges: Inbound PSTN CSS



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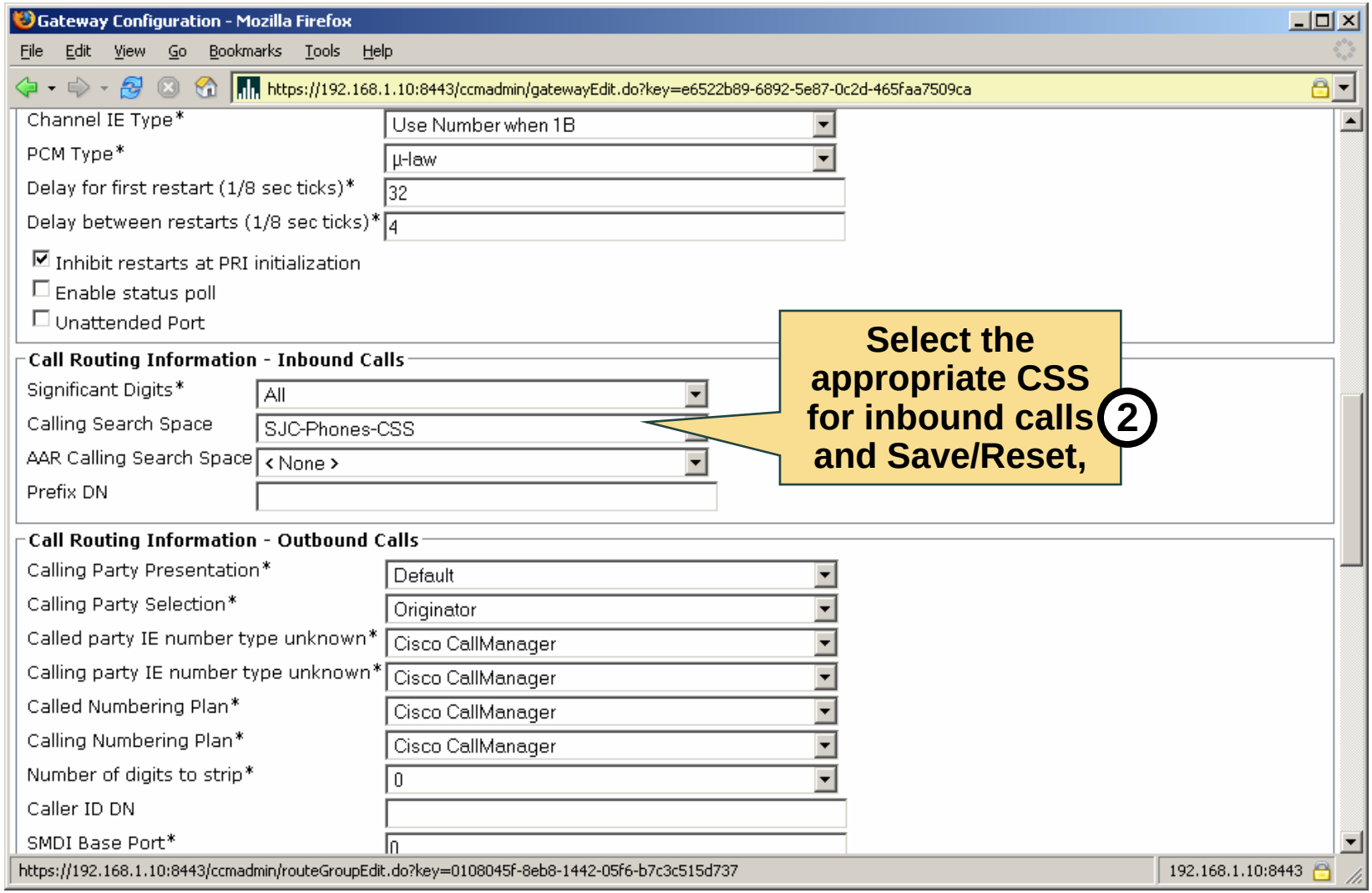
Configuring PSTN Routing and Calling Privileges: Inbound PSTN CSS



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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.

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Configuring CallManager for Multi Site Deployments

Understanding Cisco Unified CallManager Codecs and Call Admission Control

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Objectives

- **Multi-Site Characteristics**
- **Cisco Unified CallManager 5.0 Supported Codecs**
- **Understanding Regions**
- **CAC Mechanisms**
- **Understanding Locations**
- **Understanding Resource Reservation Protocol (RSVP) Based CAC**

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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.**

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Cisco Unified CallManager 5.0 Supported Codecs

Codec	Layer 3 Bandwidth	Description
G.711	80 kbps	This codec represents the default codec for all Cisco Unified CallManager calls.
G.722	80 kbps	Video endpoints typically prefer this codec.
G.723	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.
G.728	16 kbps	Video endpoints support this low-bit-rate codec.
G.729	24 kbps	The system supports this low bit-rate codec for Cisco IP Phone 7900 models.
Wideband	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.
GSM	29 kbps	The Global System for Mobile Communications (GSM) codec enables the MNET system for GSM wireless handsets to interoperate with Cisco Unified CallManager.

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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.**

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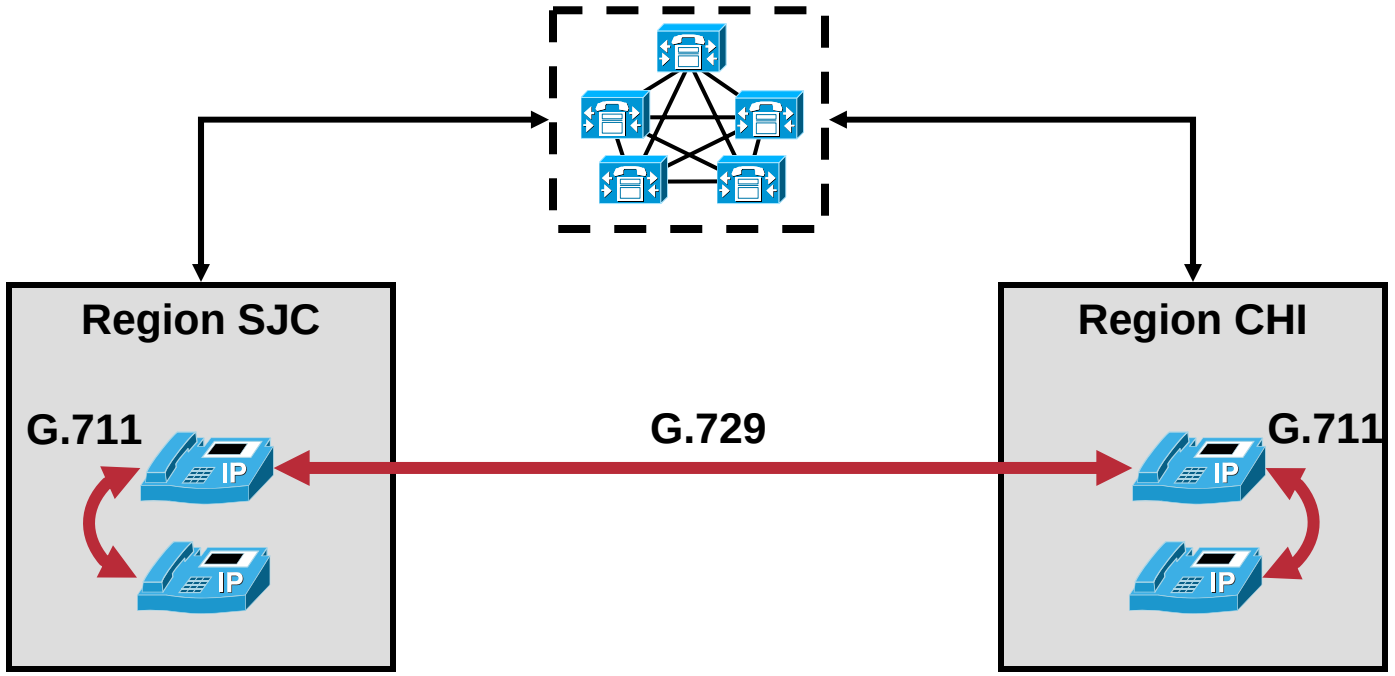
Understanding Regions



Site	Codec
With this Region (SJC)	G.711
To CHI	G.729

Site	Codec
With this Region (CHI)	G.711
To SJC	G.729

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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**

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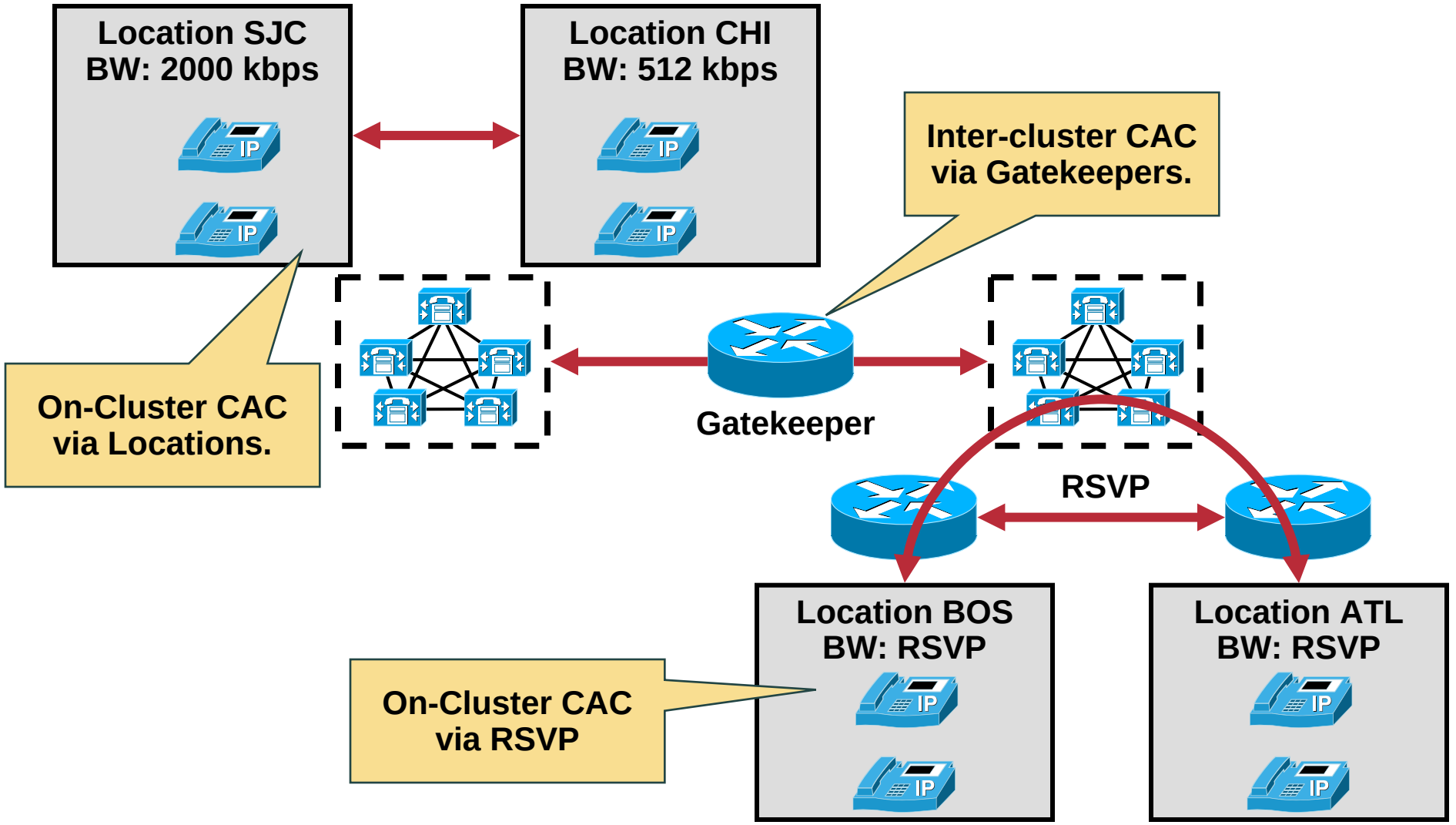
CAC Mechanisms



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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.**

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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.**

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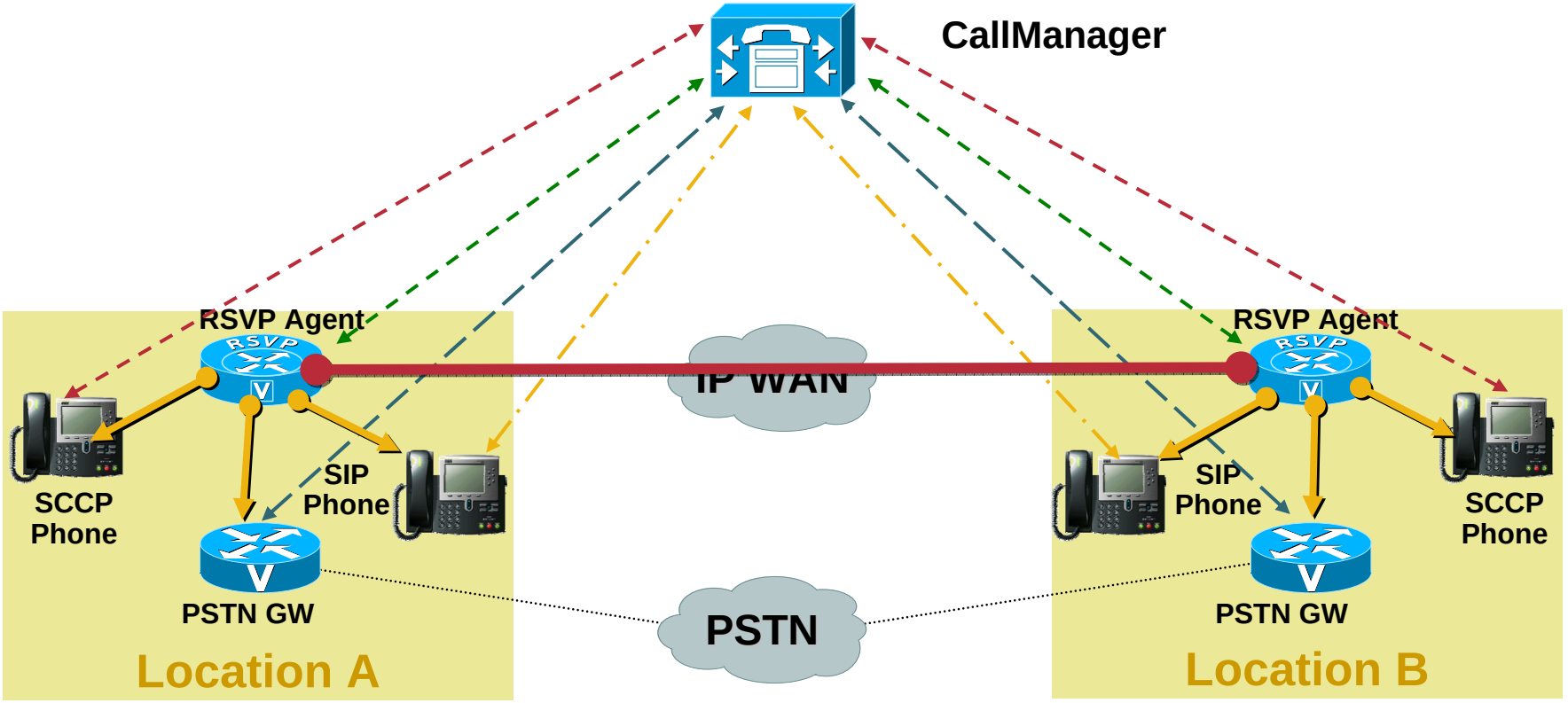
Understanding RSVP-Based CAC



SCCP

Media Resource Control

MGCP or H.323

RSVP-RTP

RTP

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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.

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Configuring CallManager for Multi Site Deployments

Configuring Regions, Locations and RSVP-Based CAC

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Objectives

- Regions Configuration
- Locations Configuration
- RSVP Concepts
- RSVP-Based CAC Configuration on Cisco CallManager
- RSVP-Based CAC Configuration on Gateways

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Regions Configuration

Region configuration steps:

1. **Add regions and specify codecs.**
2. **Assign region to device pools.**

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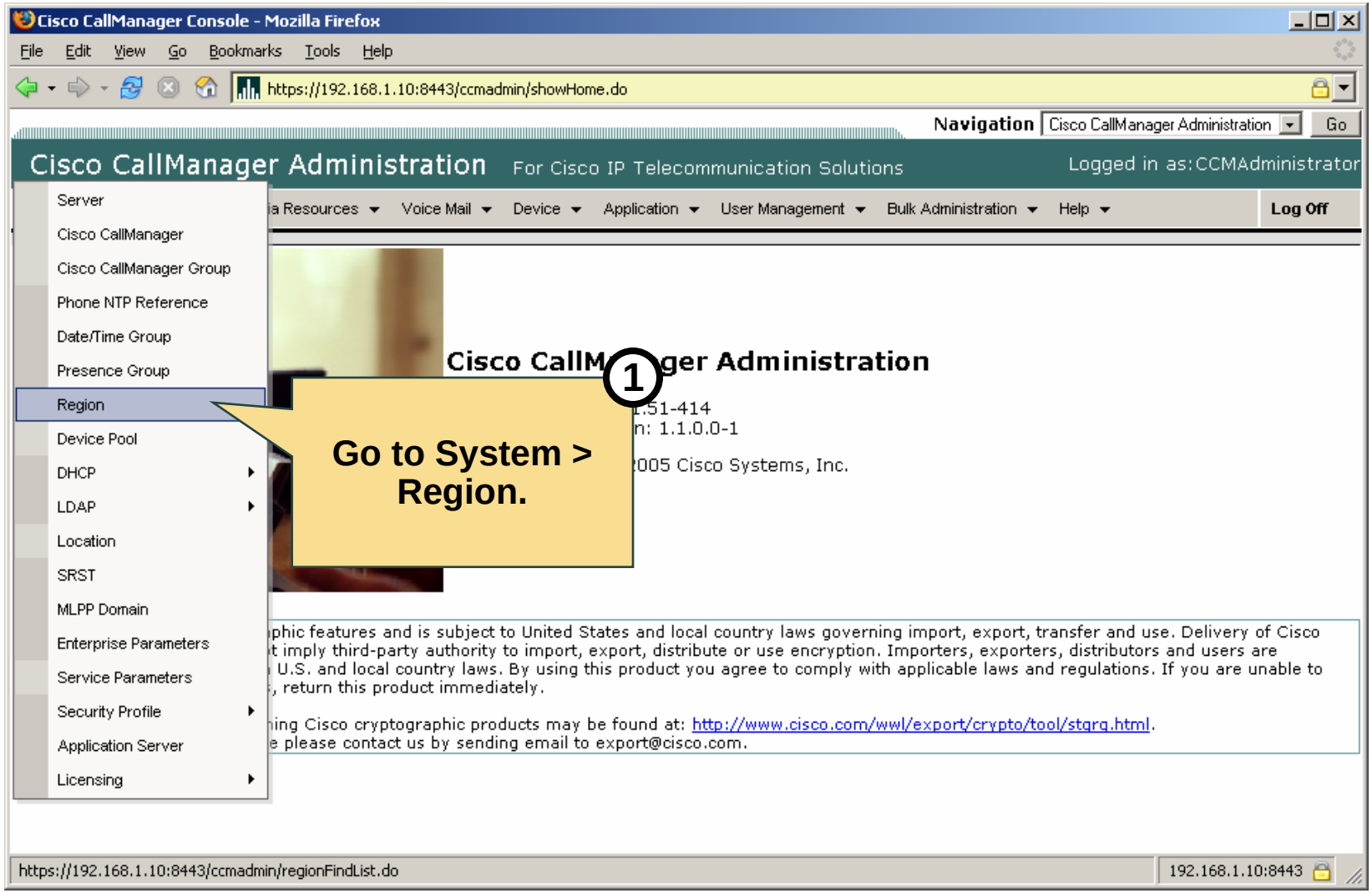
Regions Configuration



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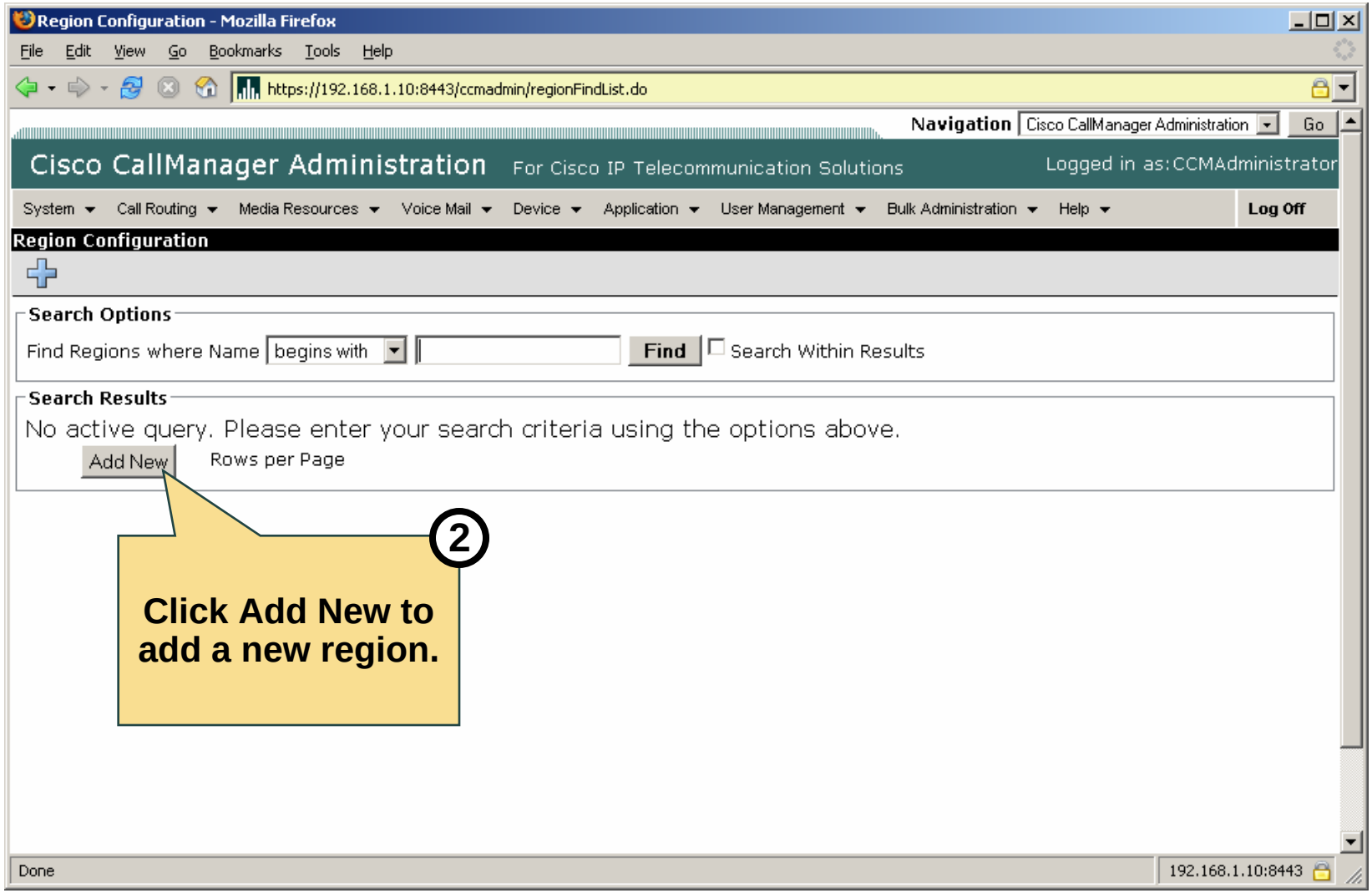
Regions Configuration



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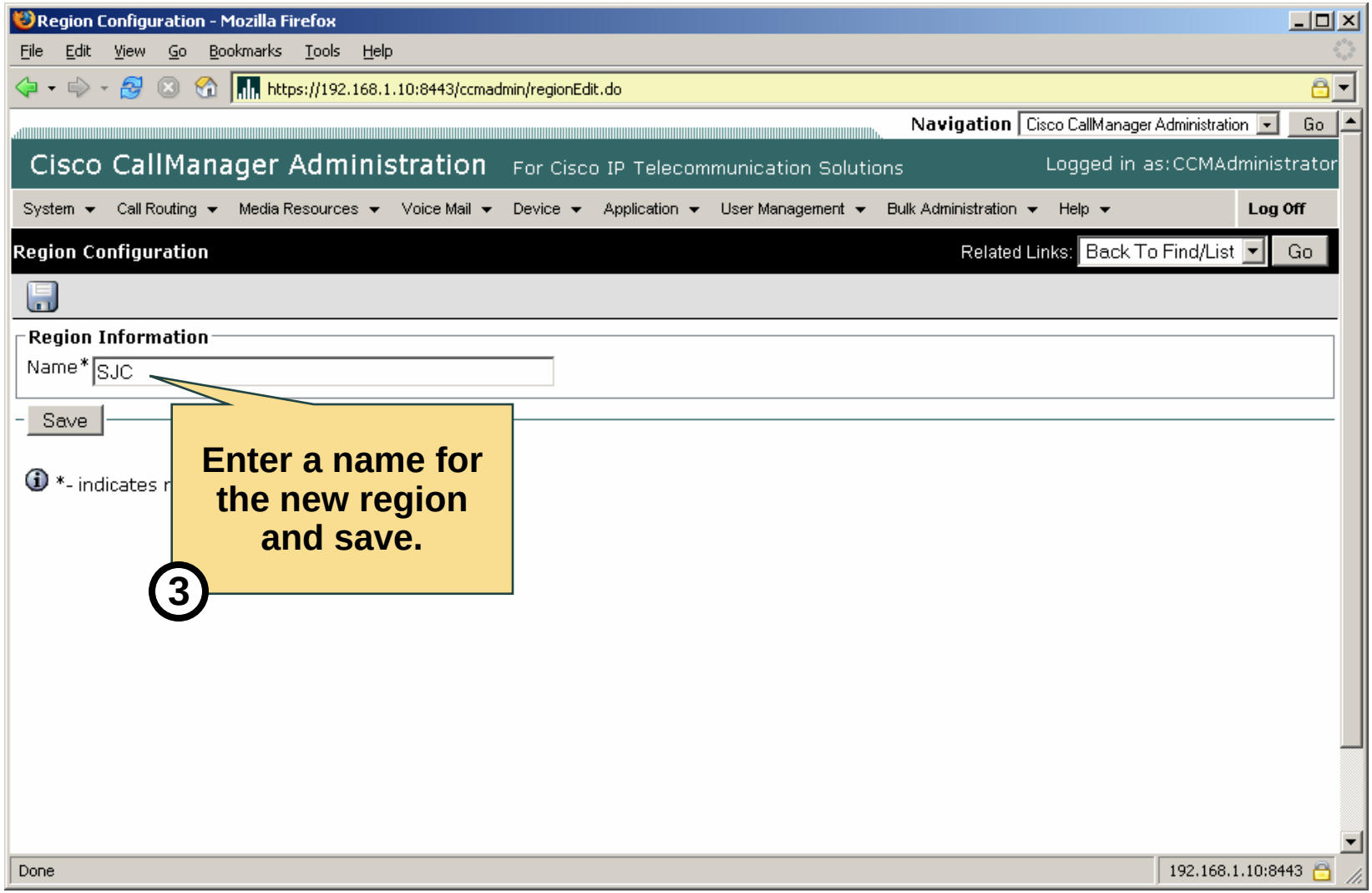
Regions Configuration



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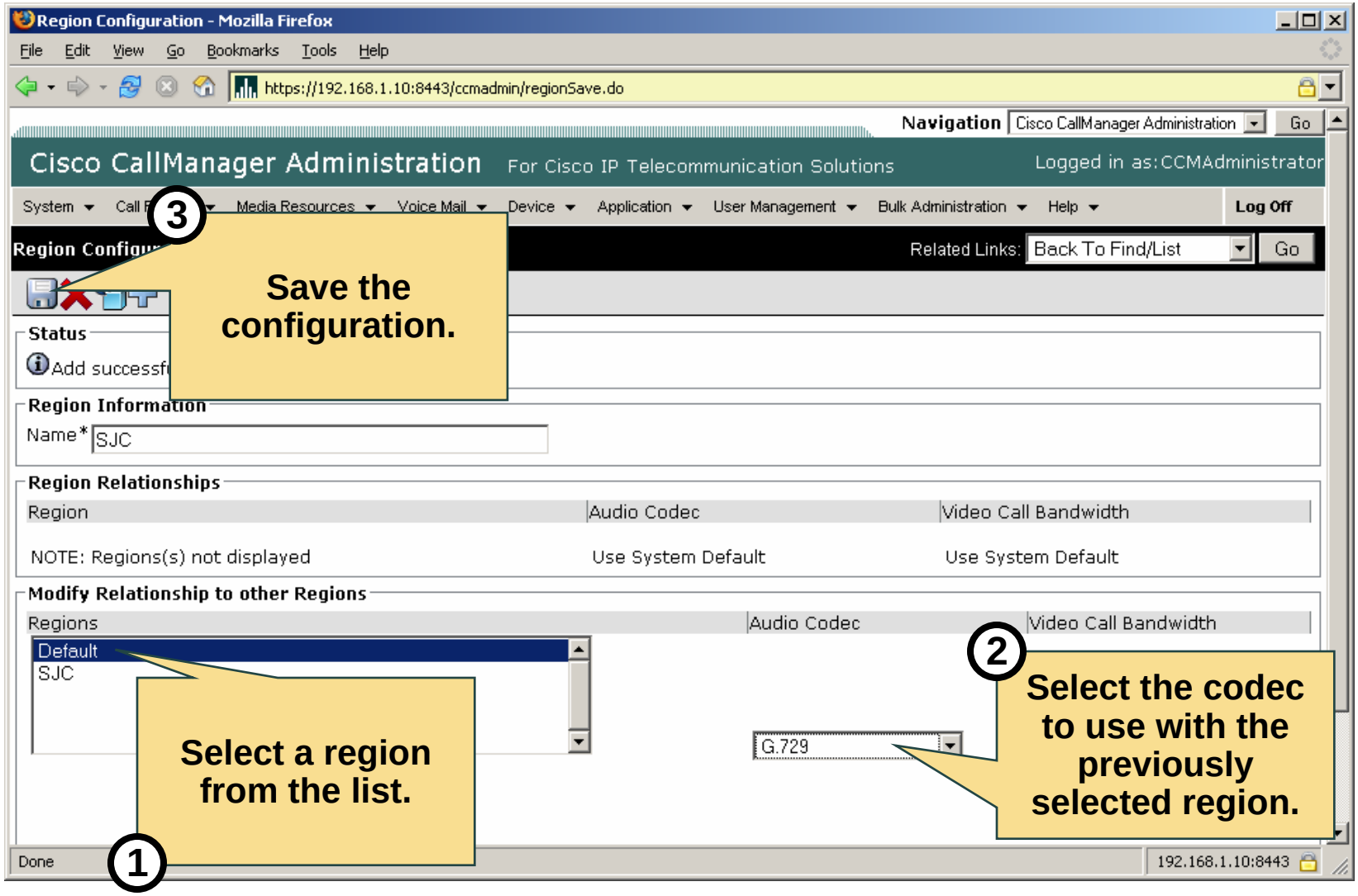

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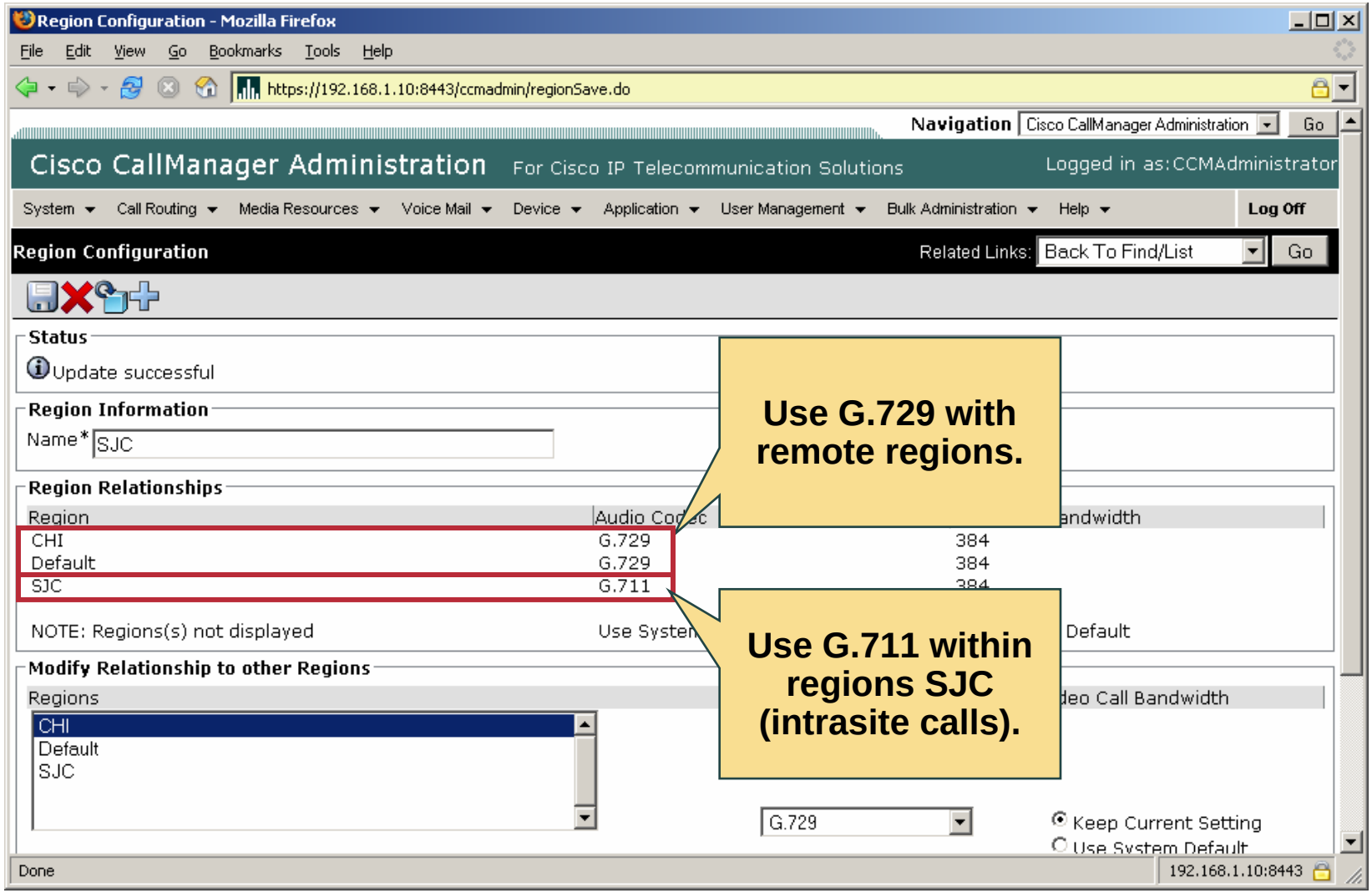

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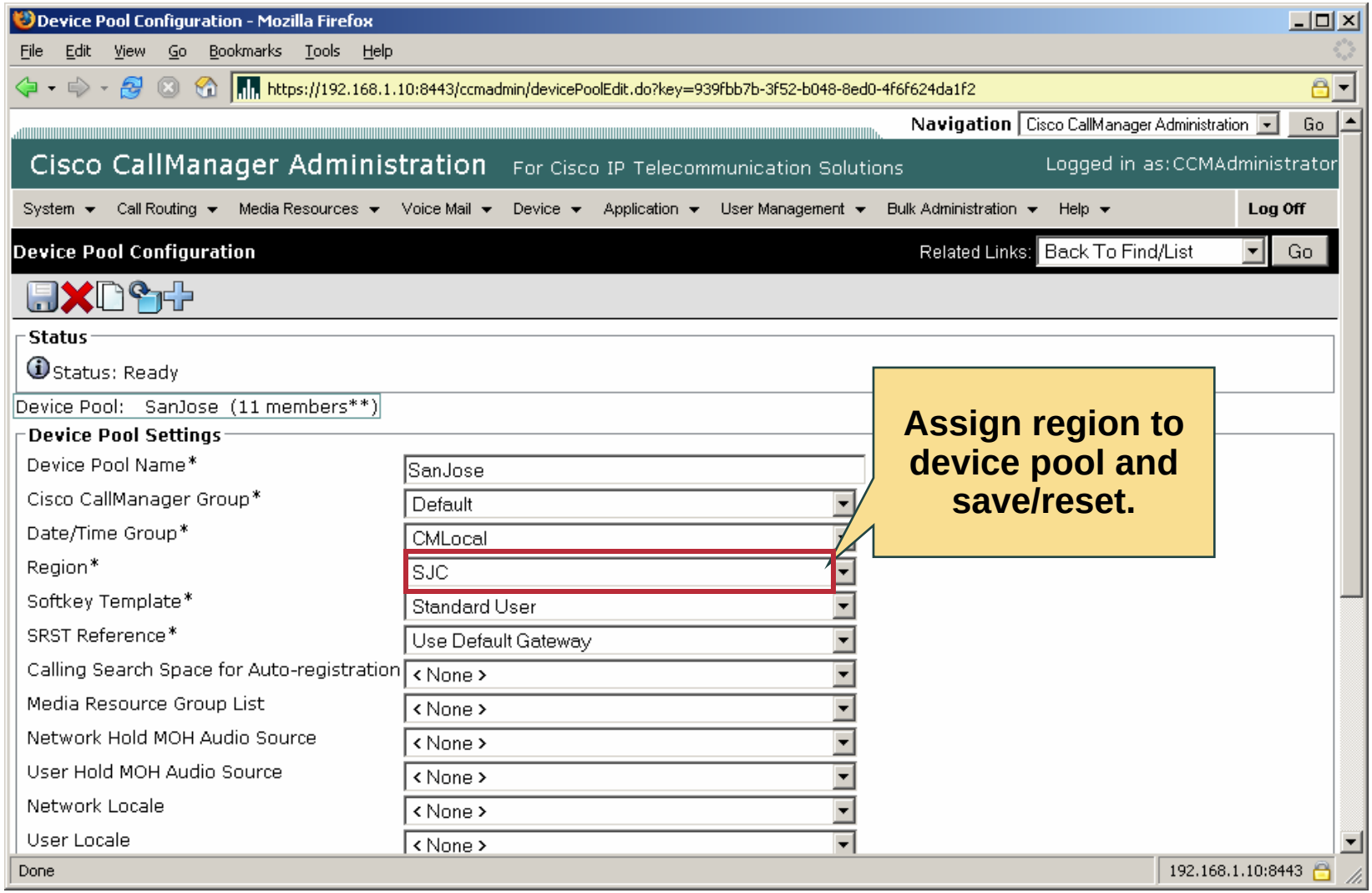
Regions Configuration



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Locations Configuration

Locations configuration steps:

1. Add locations and configure bandwidth.
2. Assign location to device.

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Locations Configuration

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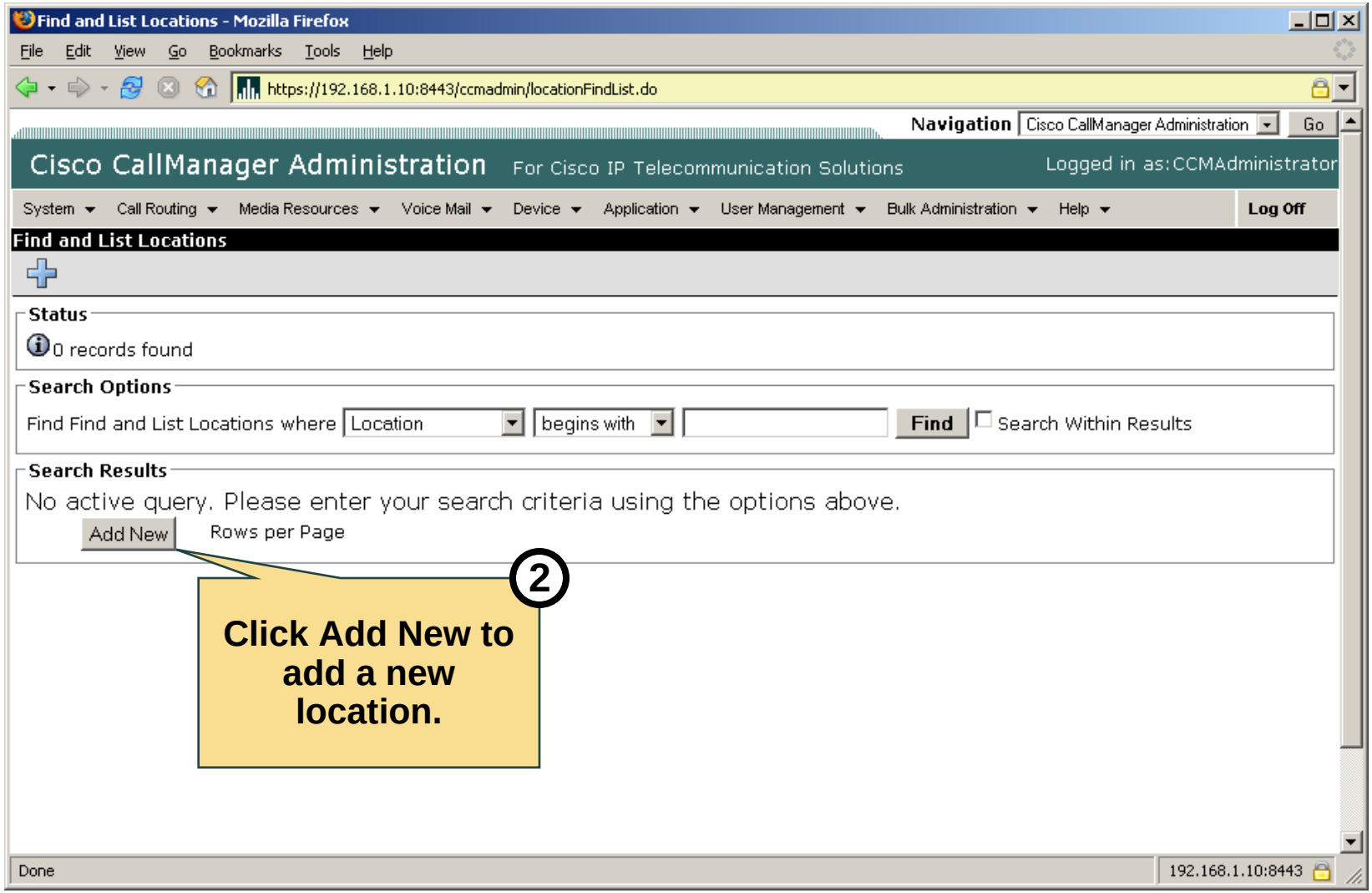
Locations Configuration



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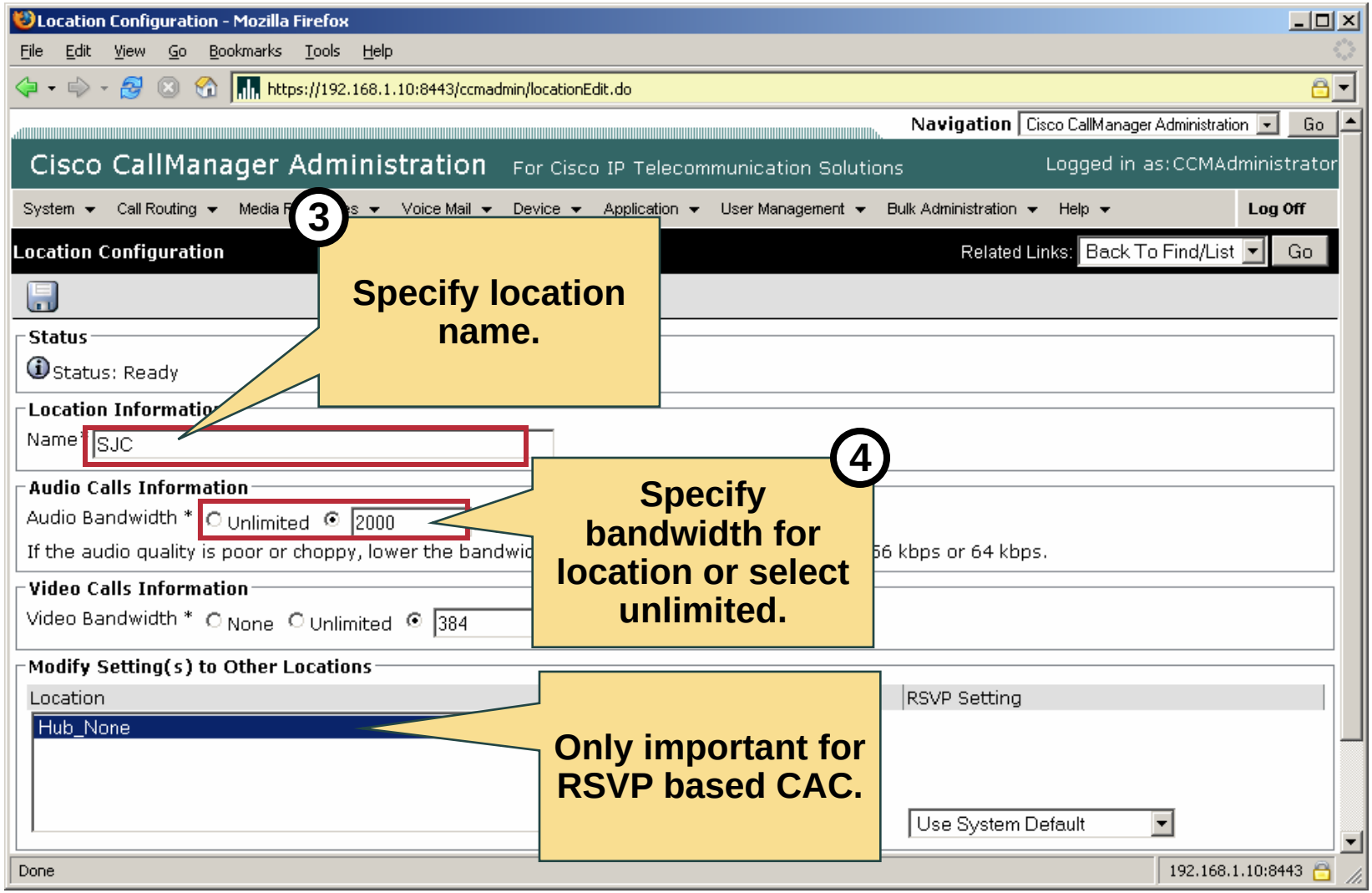
Locations Configuration



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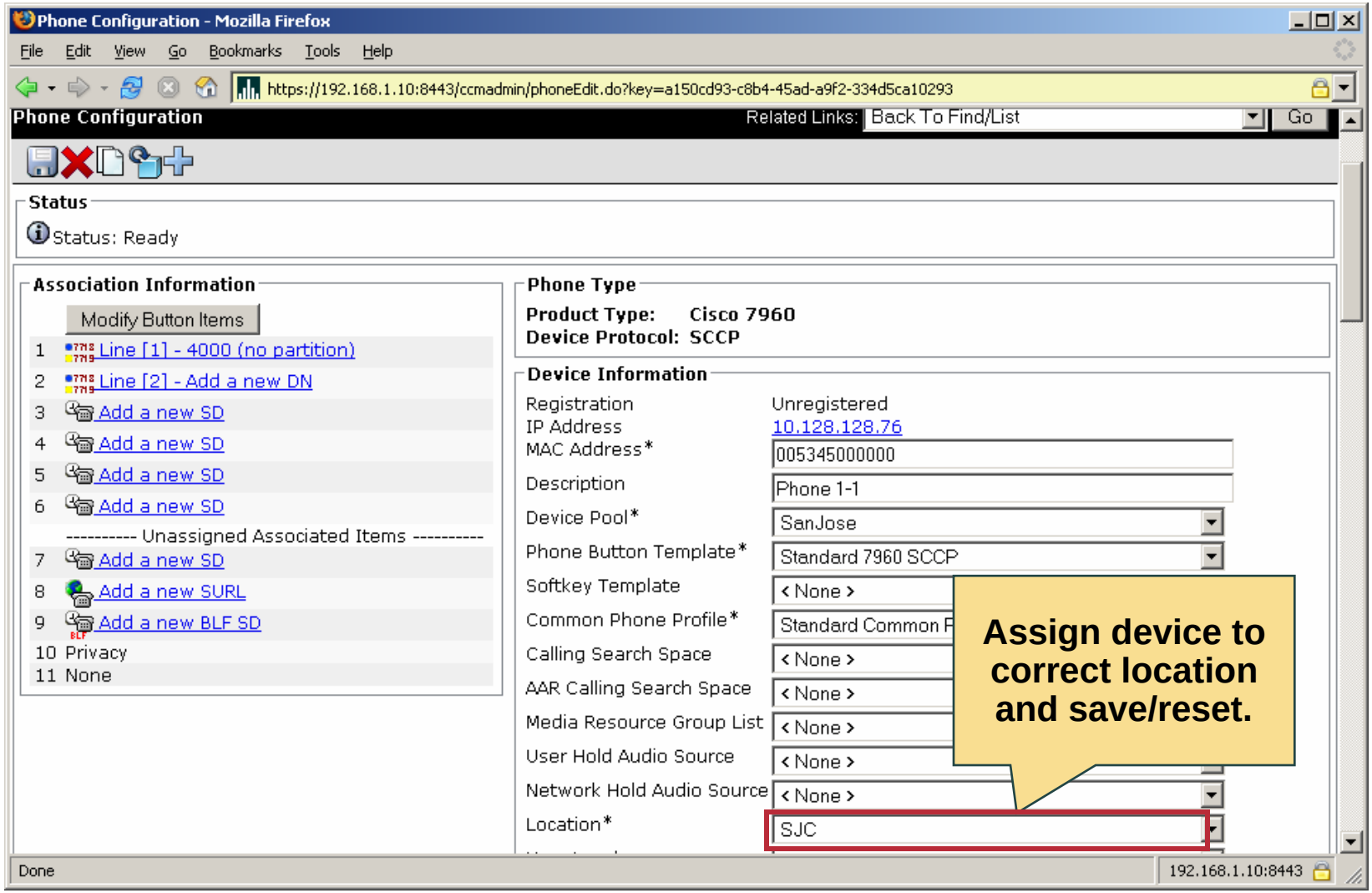
Locations Configuration



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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**

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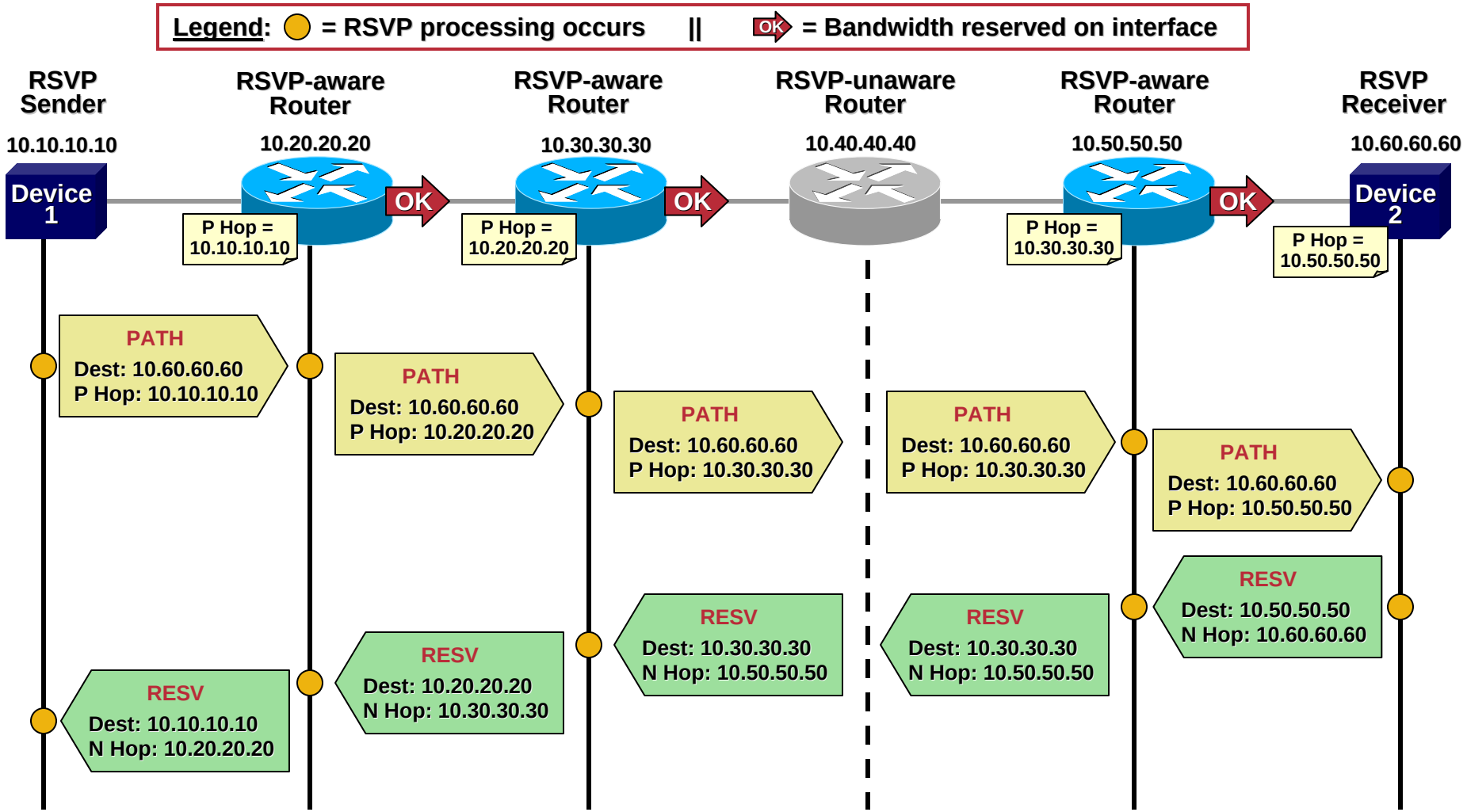
RSVP Concepts



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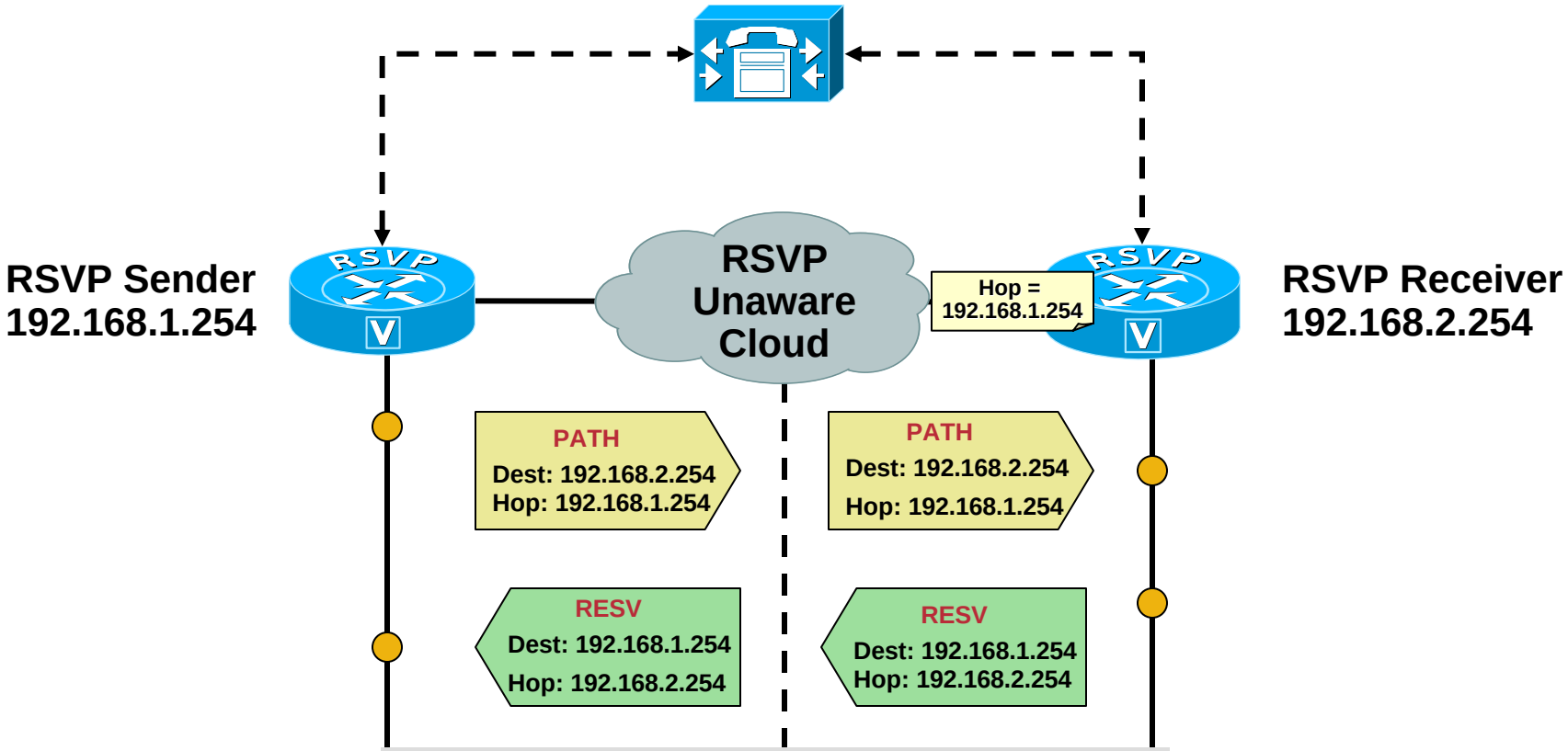
RSVP Concepts



	Reservation during setup	Reservation during call
G.711	96 kbps	80 kbps
G.729	40 kbps	24 kbps

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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.**

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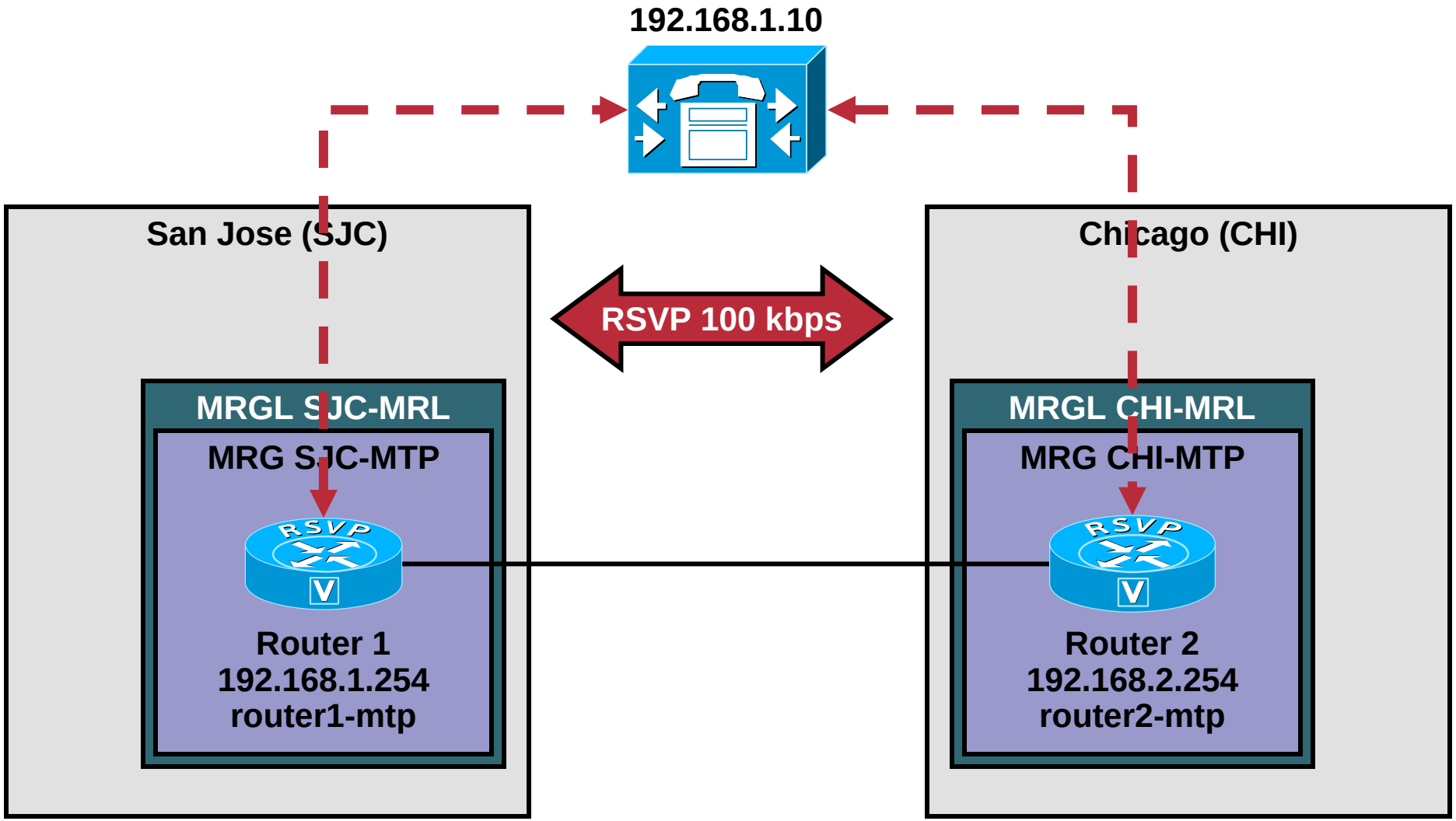
RSVP-Based CAC Configuration Steps on Cisco Unified CallManager



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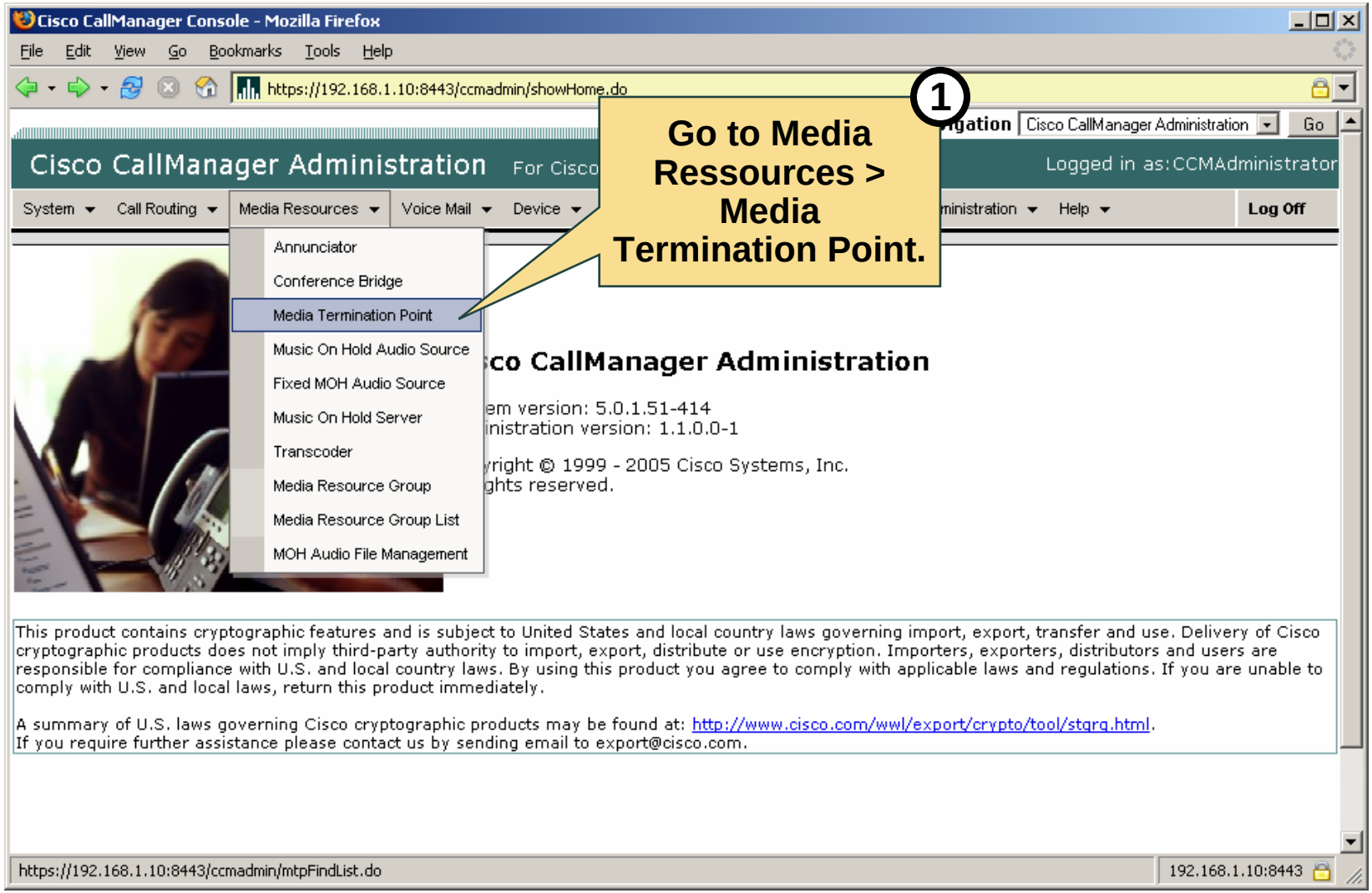
Media Termination Point Configuration



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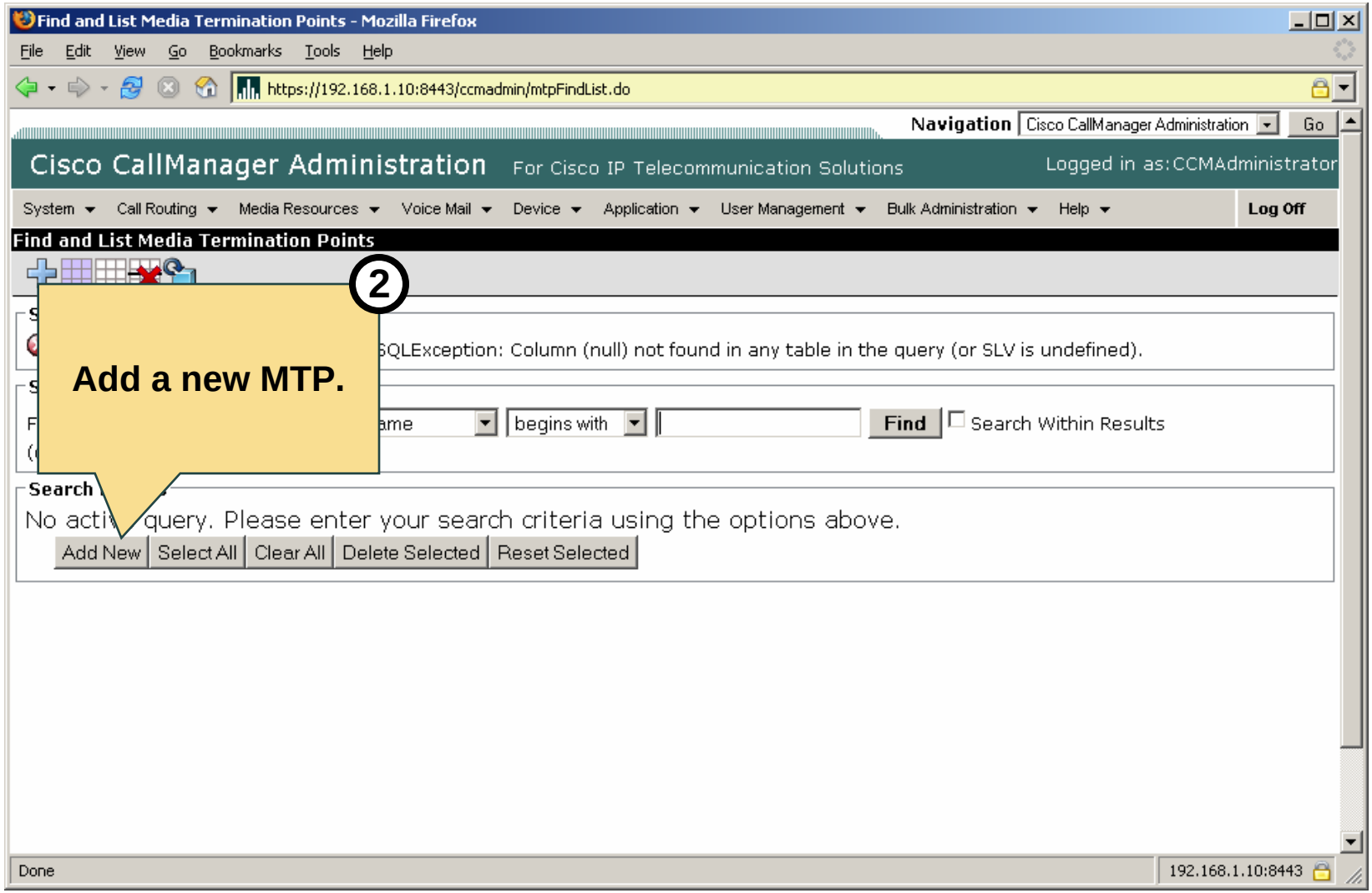
Media Termination Point Configuration



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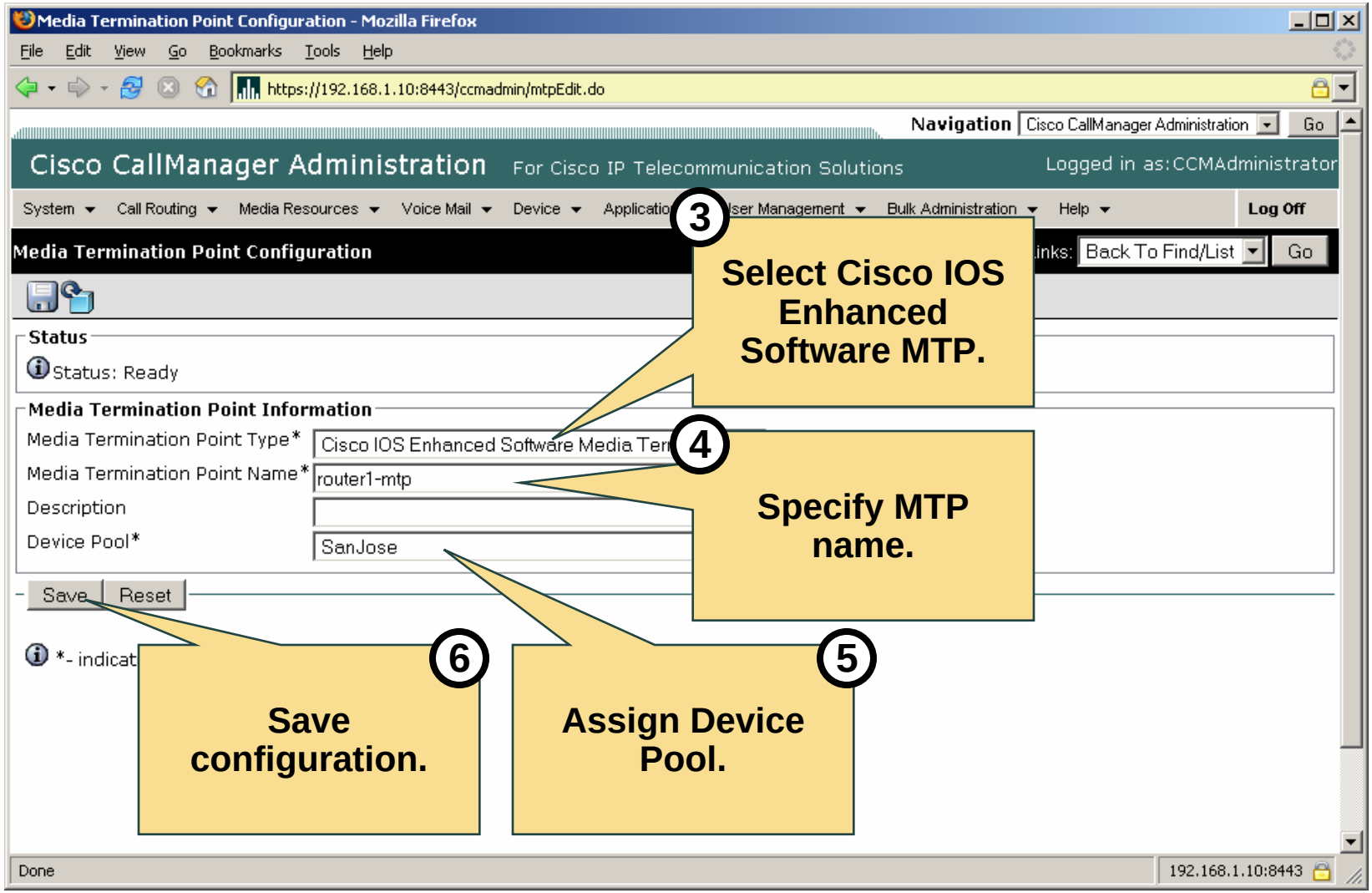
Media Termination Point Configuration



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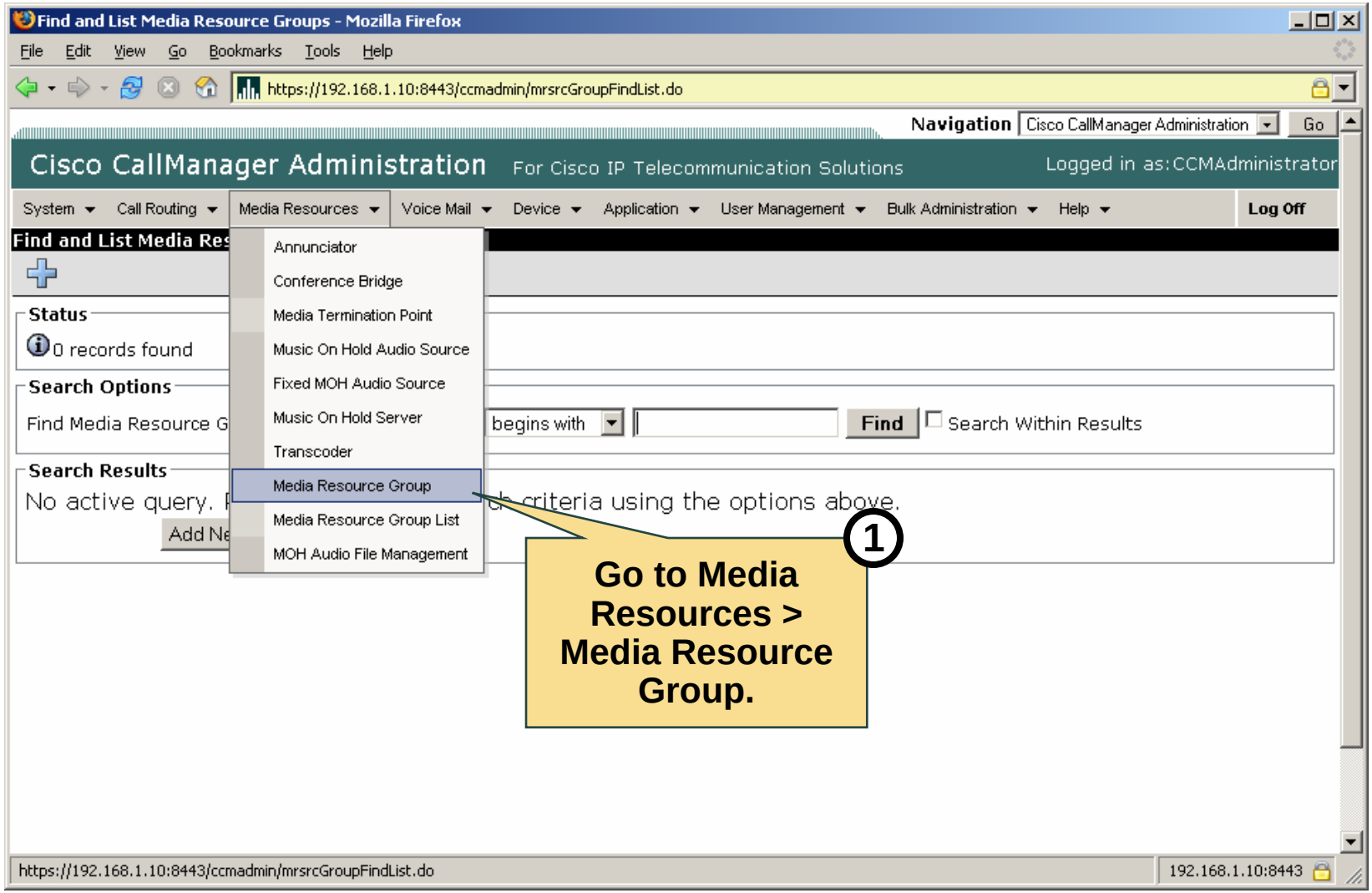
Media Resource Group and List Configuration



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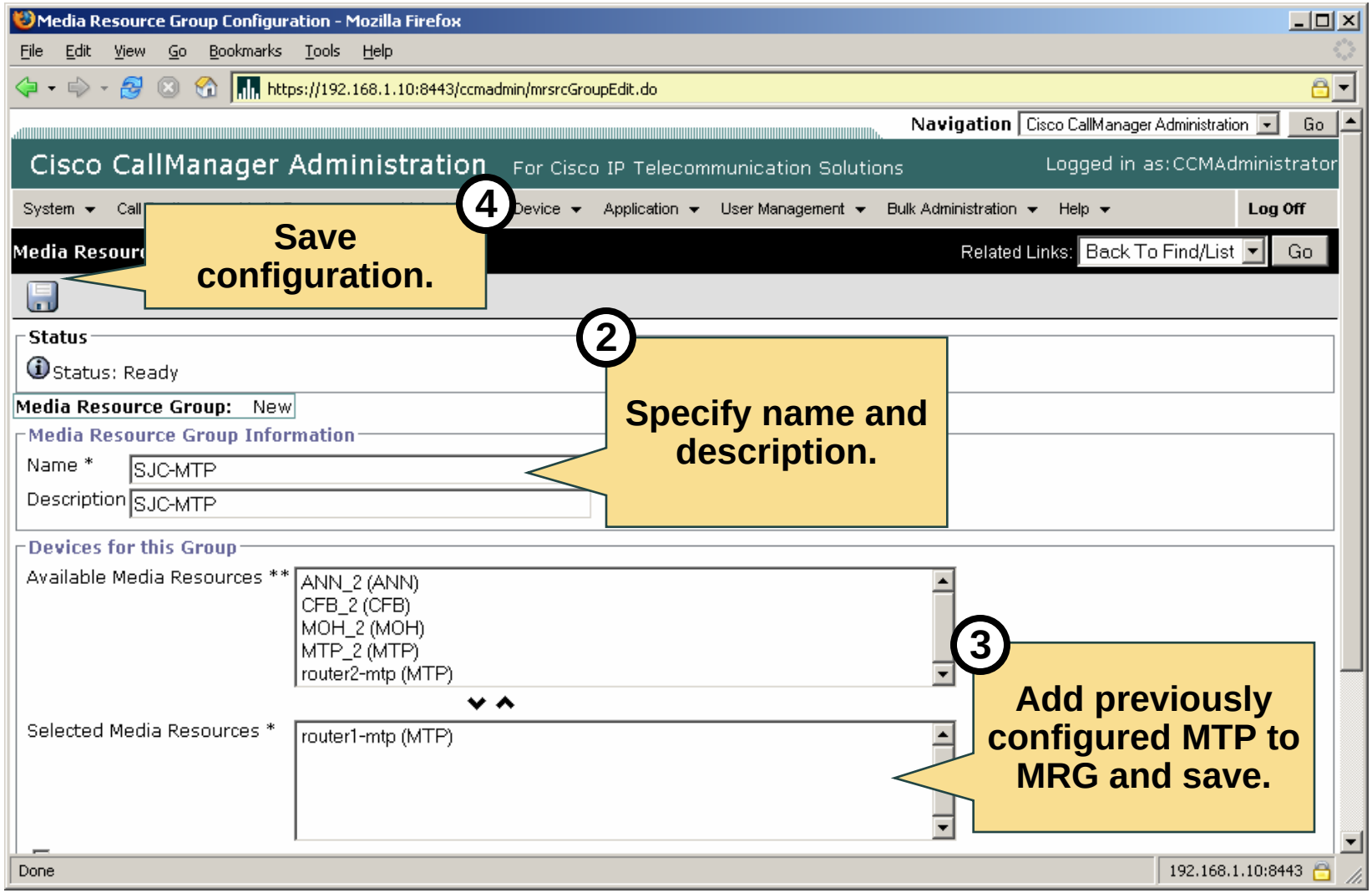
Media Resource Group and List Configuration



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Media Resource Group and List Configuration

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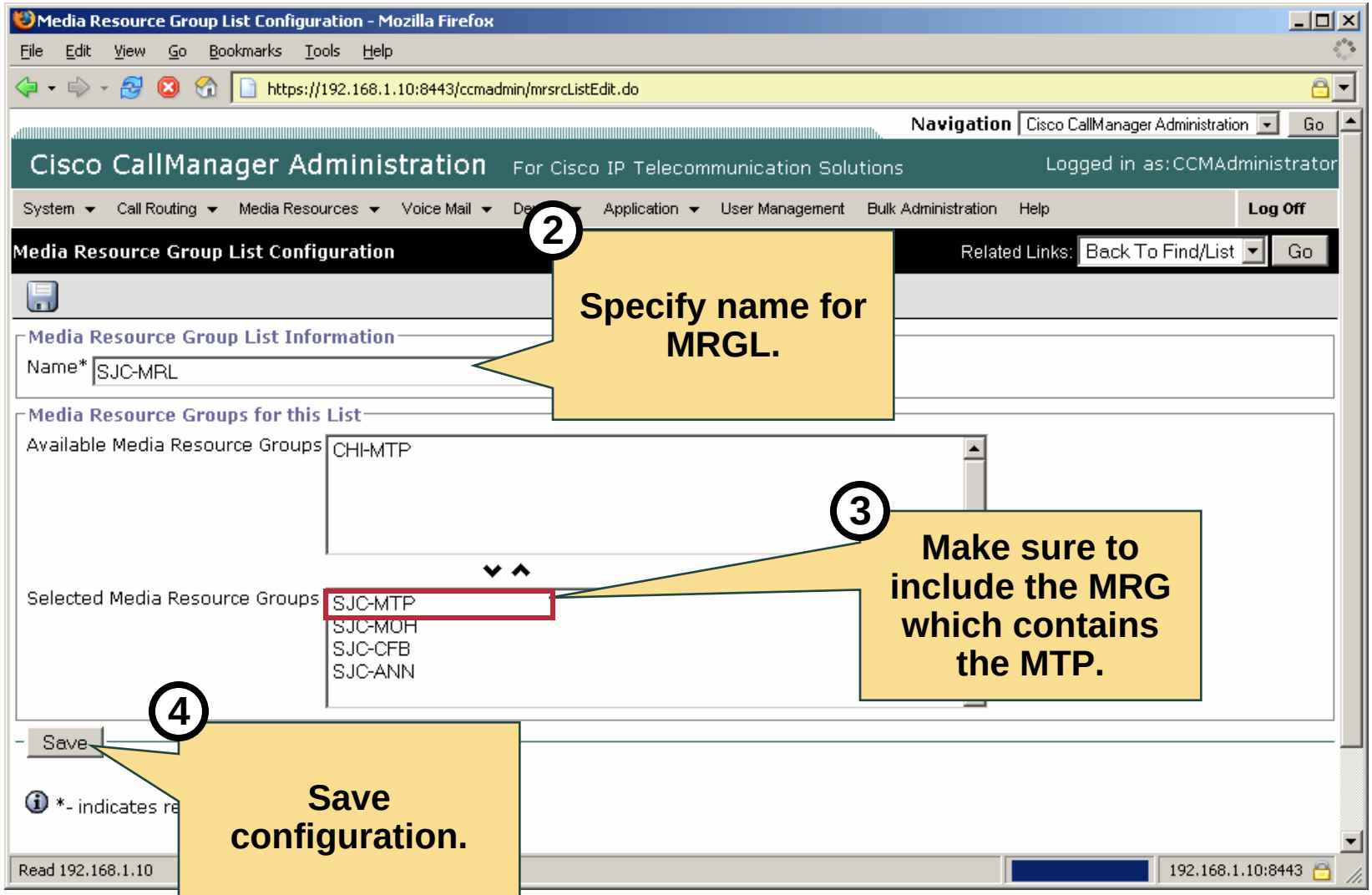
Media Resource Group and List Configuration



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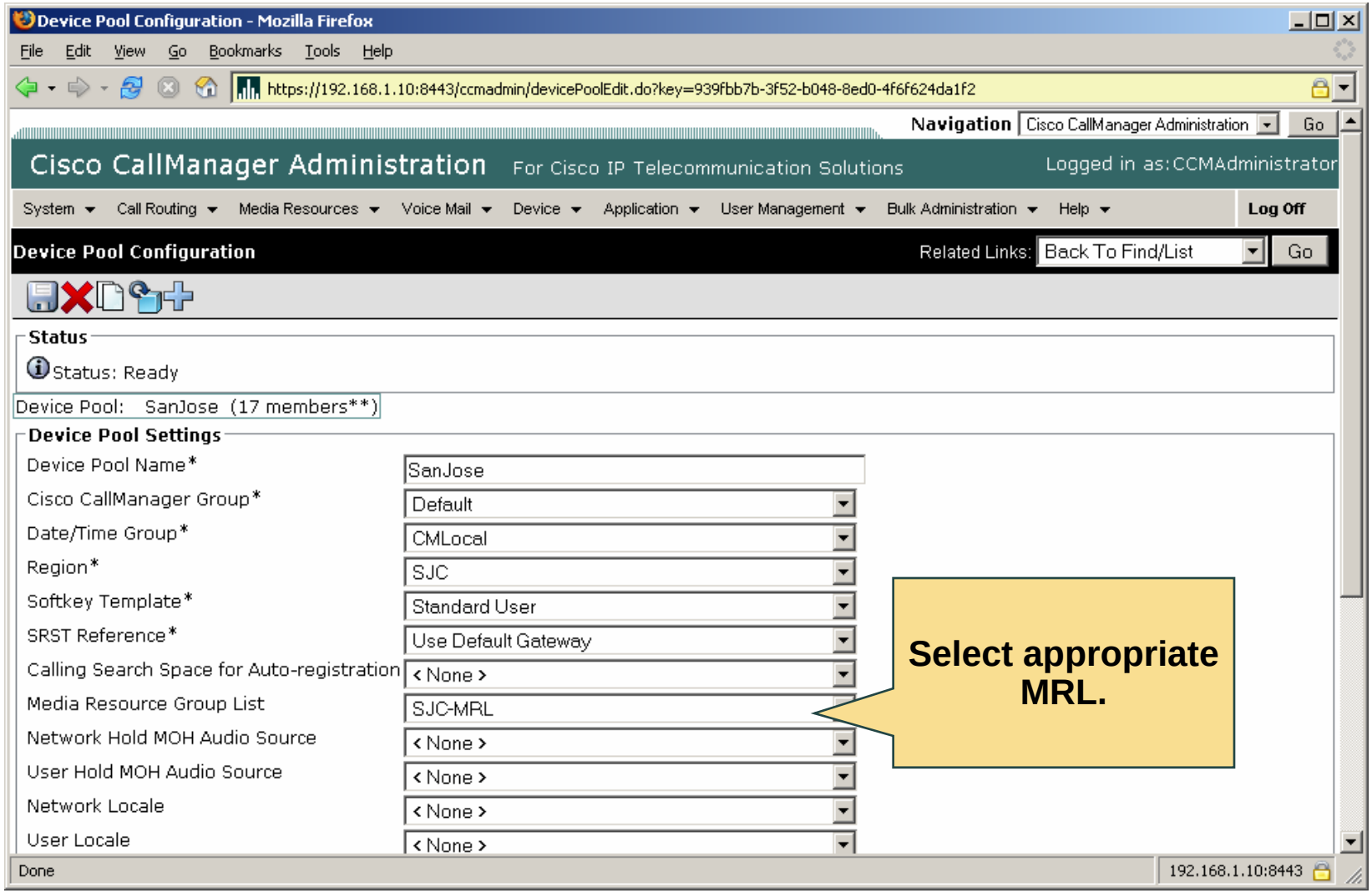
Media Resource Group and List Configuration



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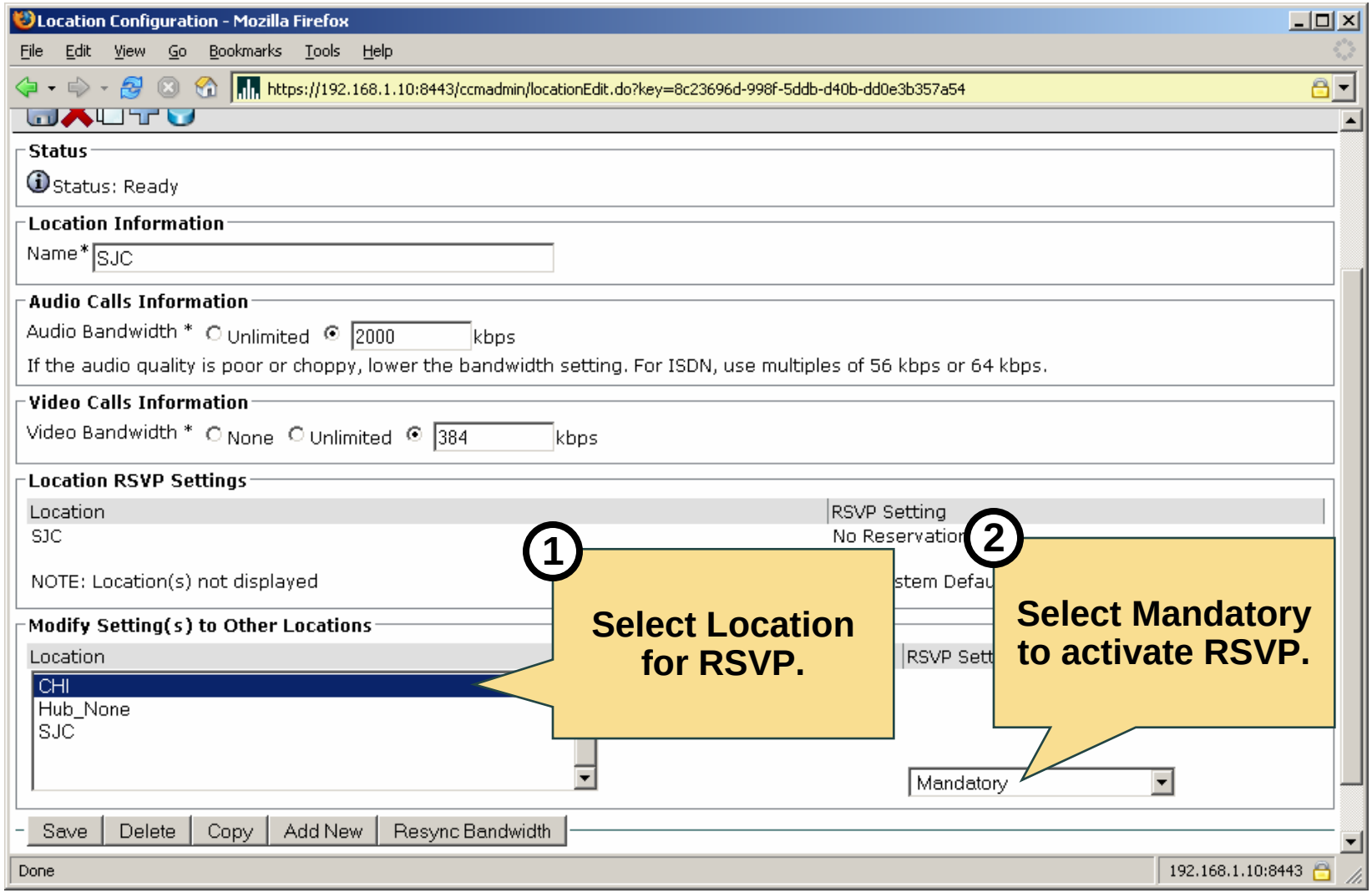
RSVP-Based CAC Configuration on Cisco Unified CallManager



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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)

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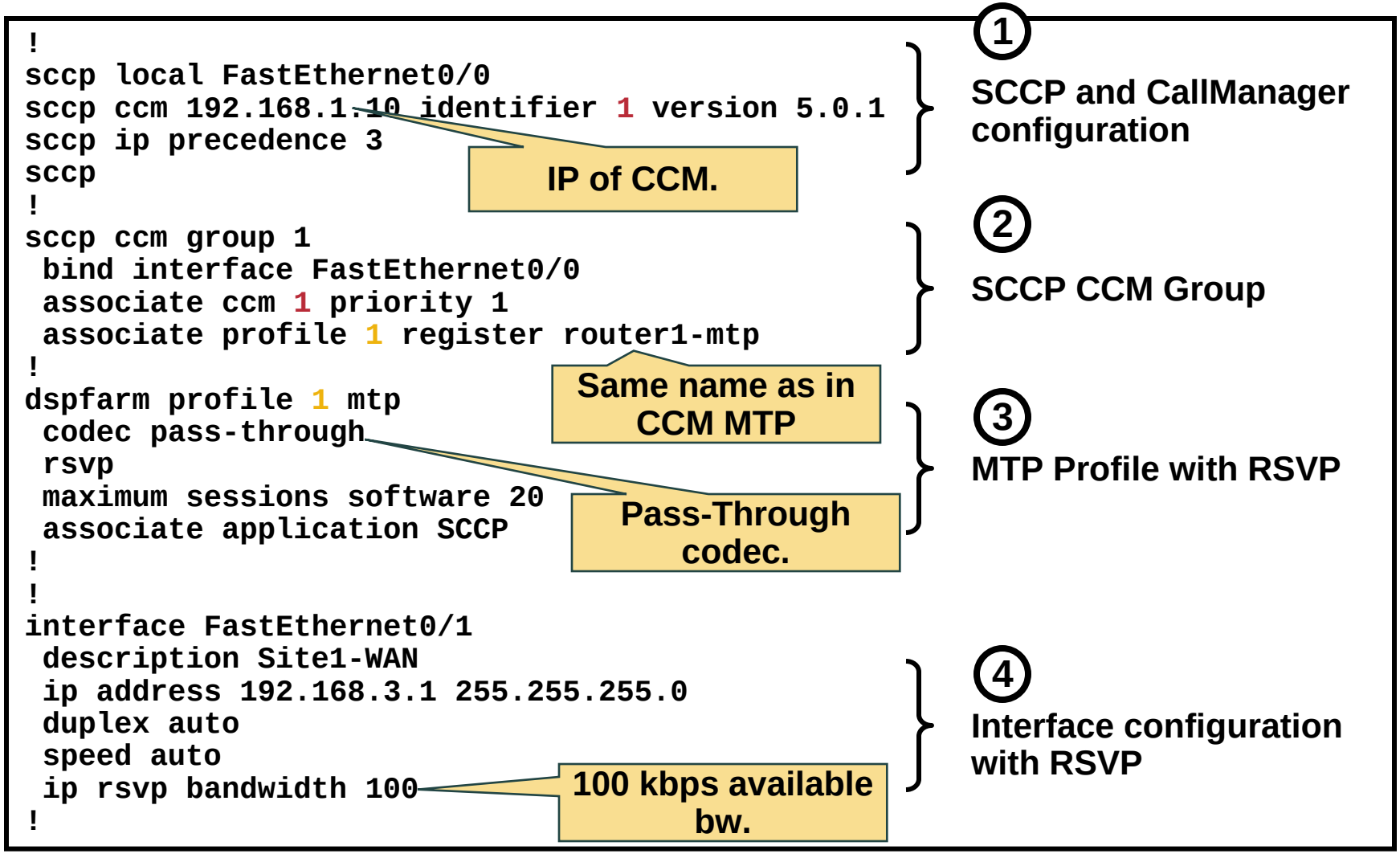
RSVP-Based CAC Configuration on Gateways



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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.

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Configuring CallManager for Multi Site Deployments

Configuring Survivable Remote Site Telephony (SRST)

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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

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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**

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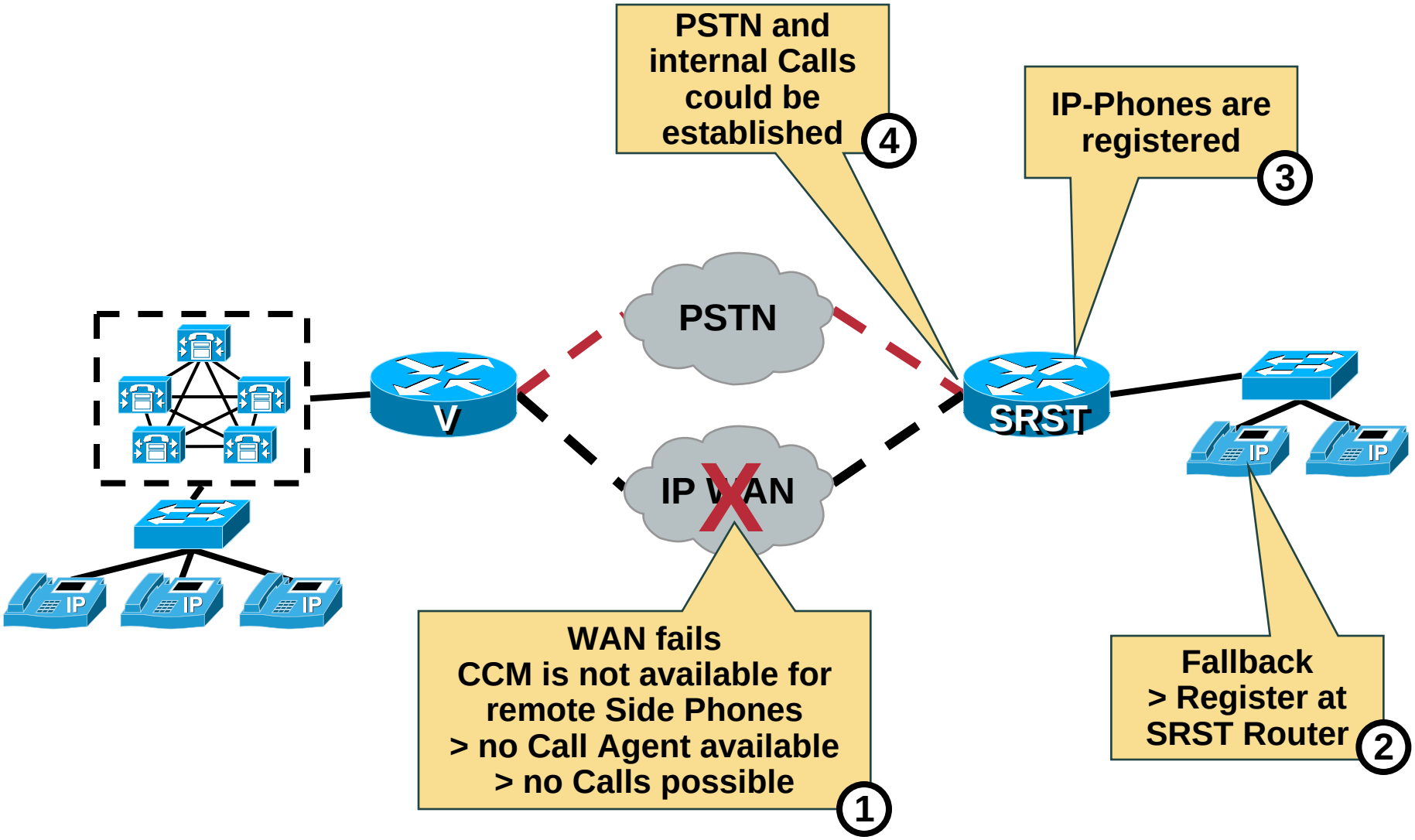
SRST Functionality



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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**

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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

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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

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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

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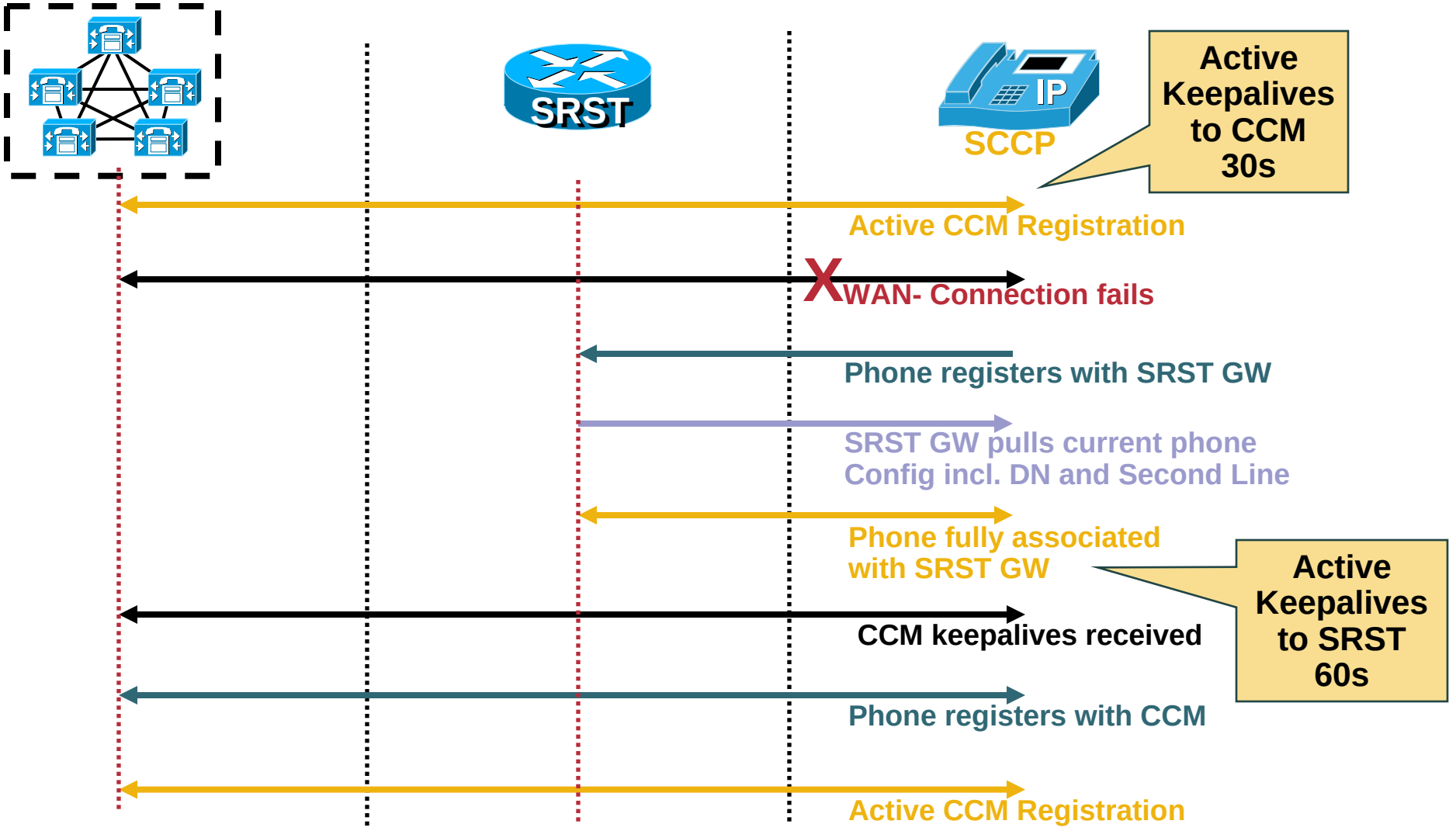
SRST Phone Registration Process



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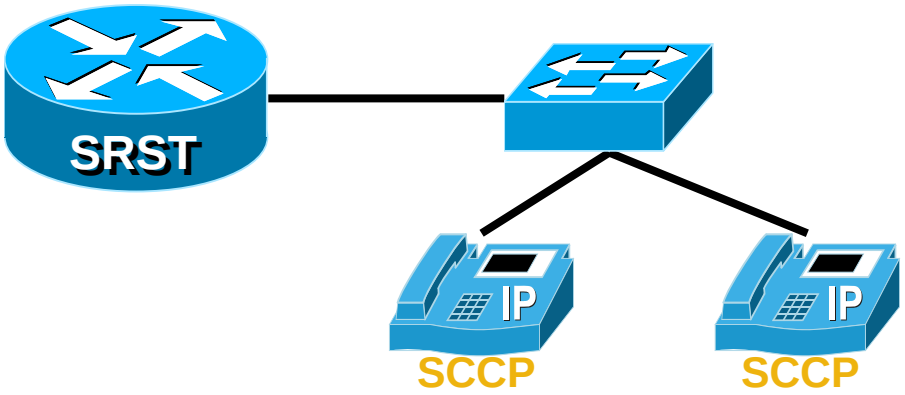
Configuring SRST for SCCP Support

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



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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**

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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.**

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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)# 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)#
```

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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.**

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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
```

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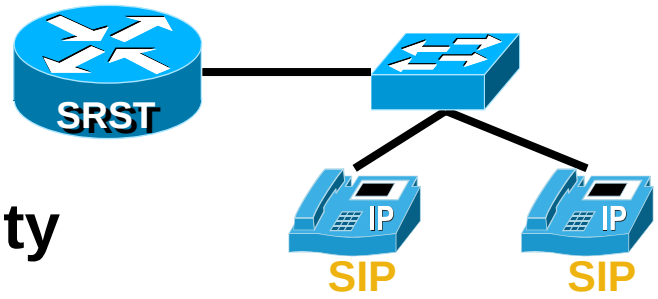
Configuring SRST for SIP Support

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



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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.

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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**

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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.**

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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
!
```

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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**

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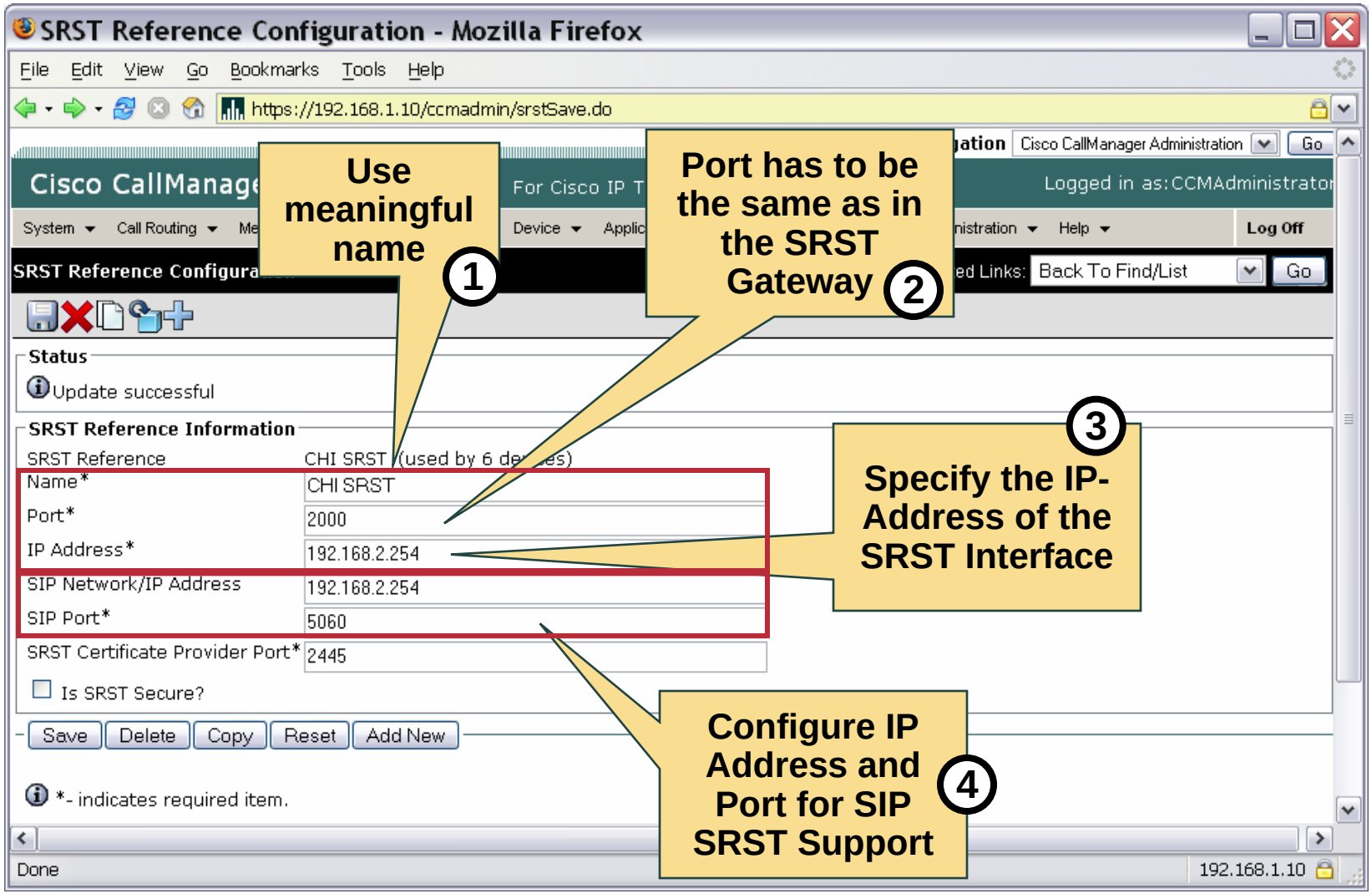
SRST Reference in Cisco Unified CallManager



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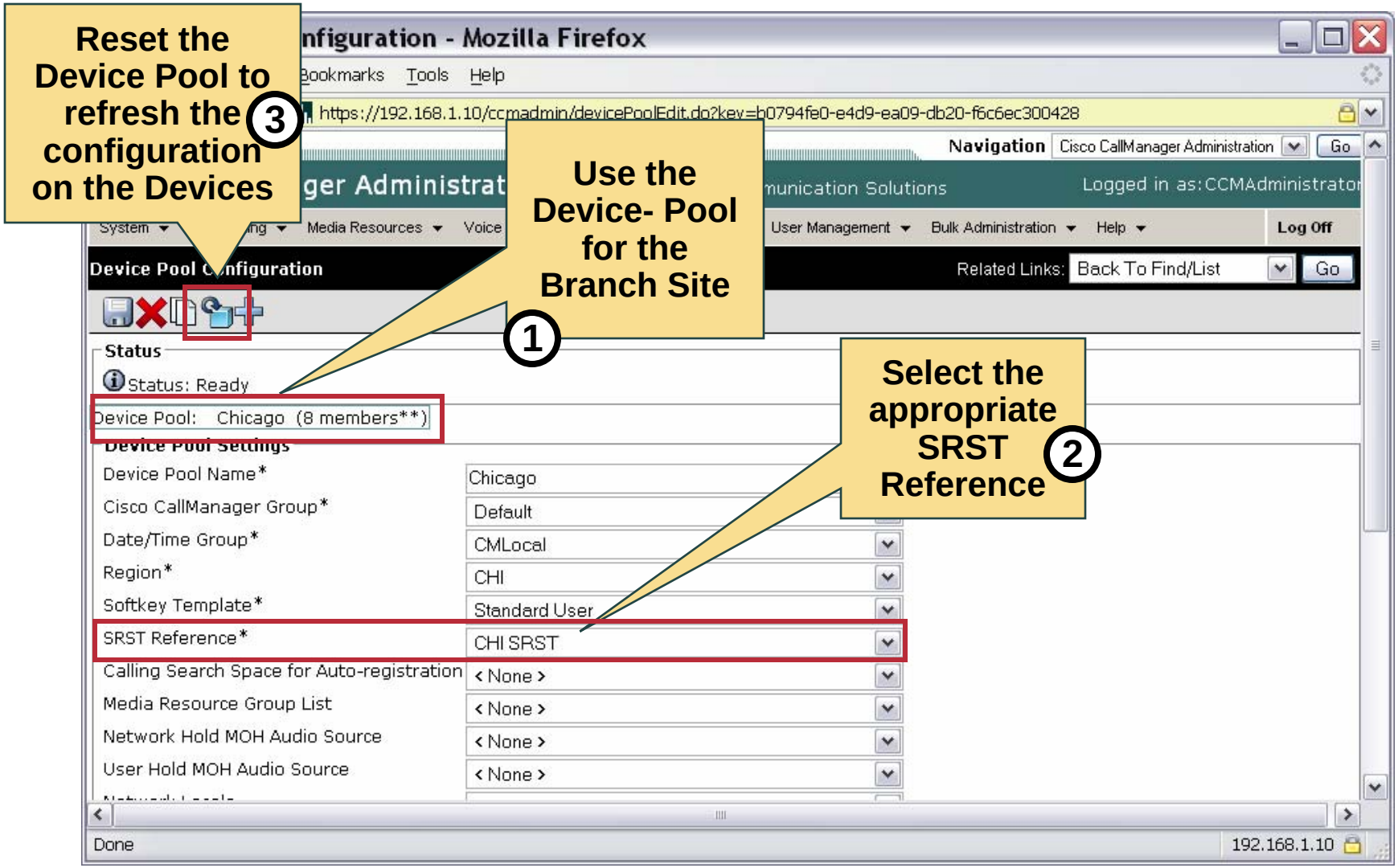
SRST Reference Association in Device Pools



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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

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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

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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
!
!
```

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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.

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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.

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Cisco Unified CallManager 5.0 Features and Applications

Introducing Cisco Unified CallManager 5.0 Applications

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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

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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**

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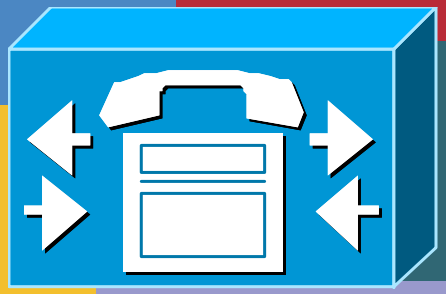
Cisco Unified CallManager 5.0 Out-of-the-Box Applications

IPMA
Manager Assistant Application to enhance productivity and cover needs of executive stuff

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

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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**

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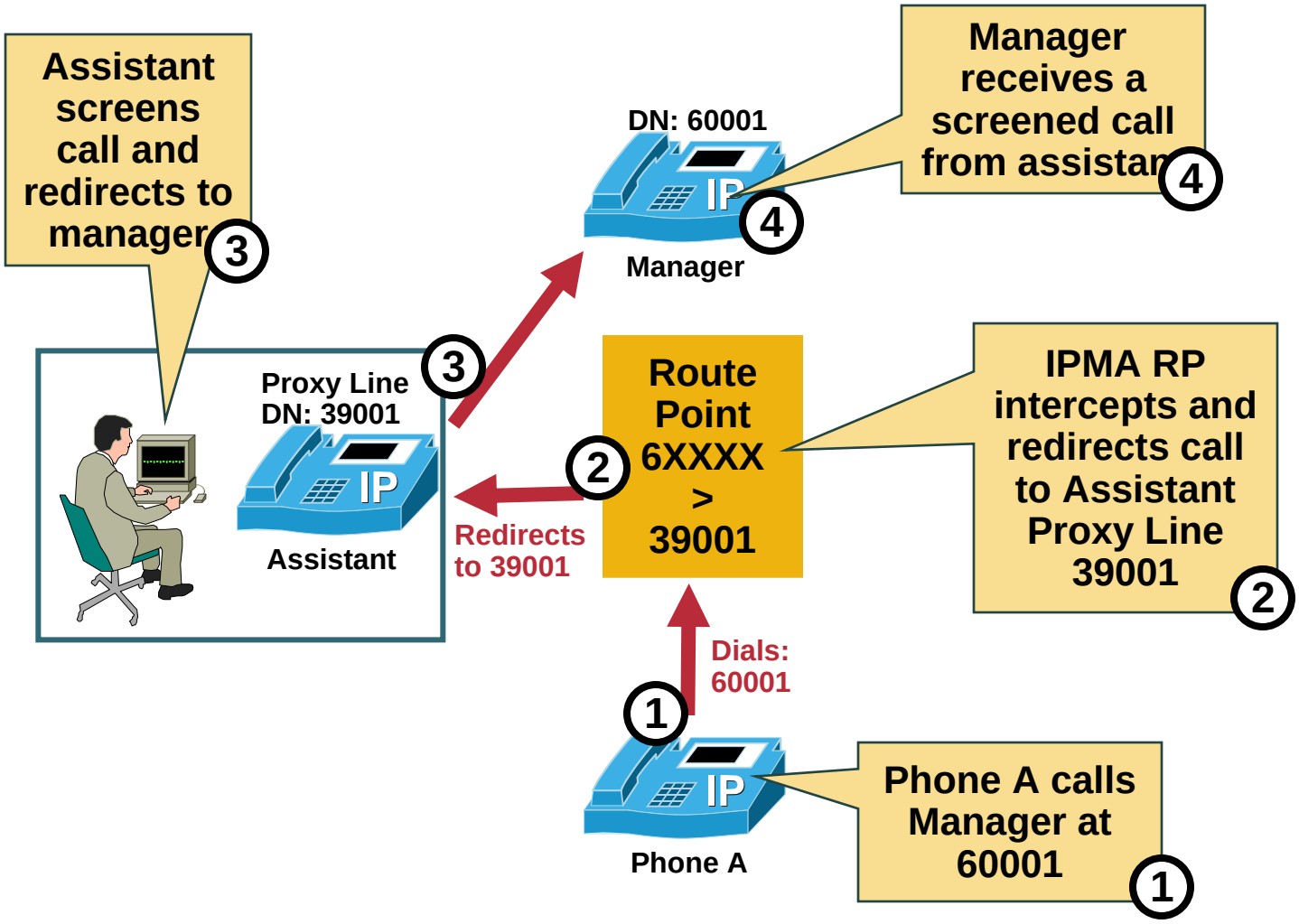
IPMA Overview: Proxy Line Mode



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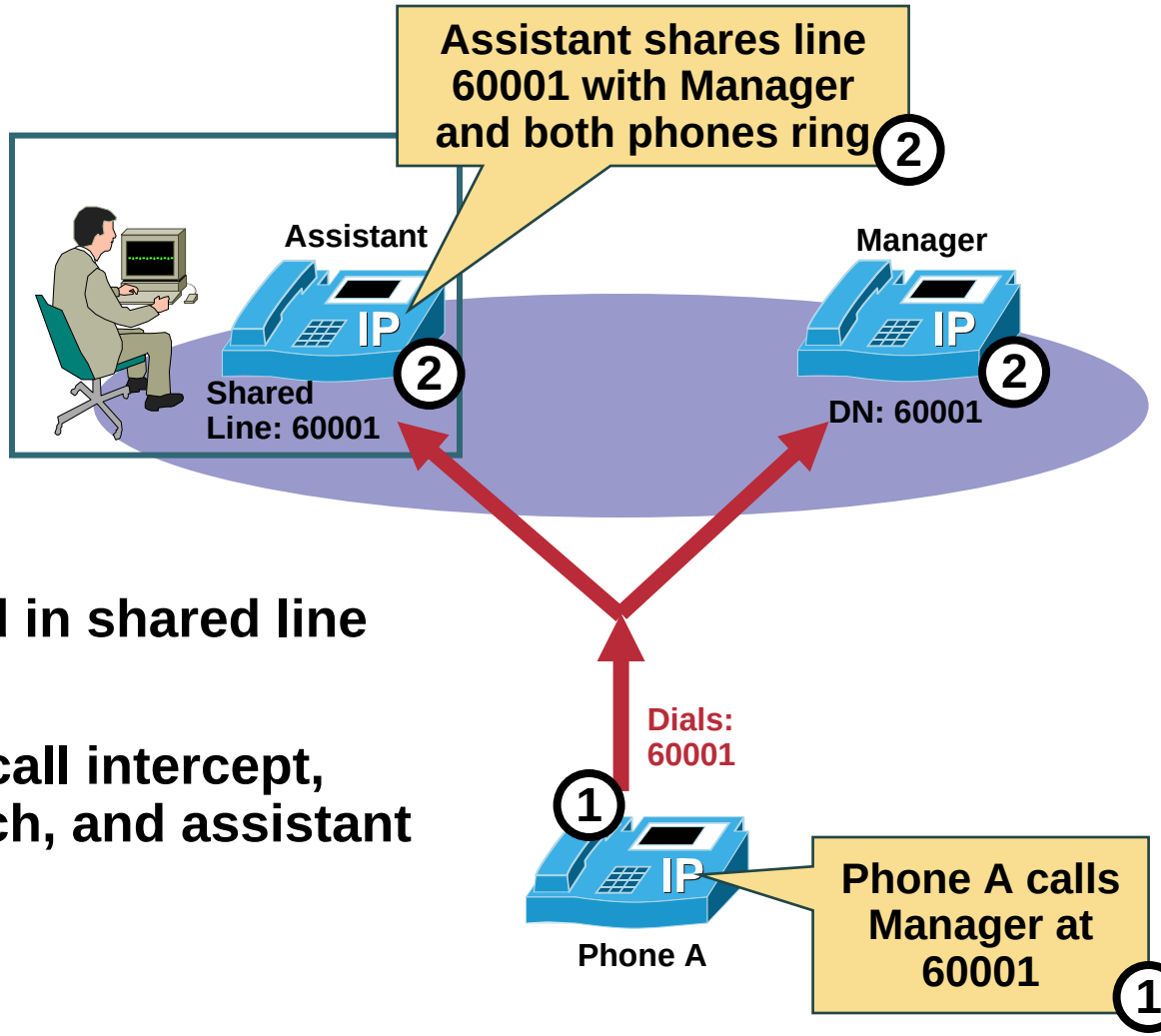
IPMA Overview: Shared Line Mode



Not supported in shared line mode:

Call filtering, call intercept, assistant watch, and assistant selection

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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

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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

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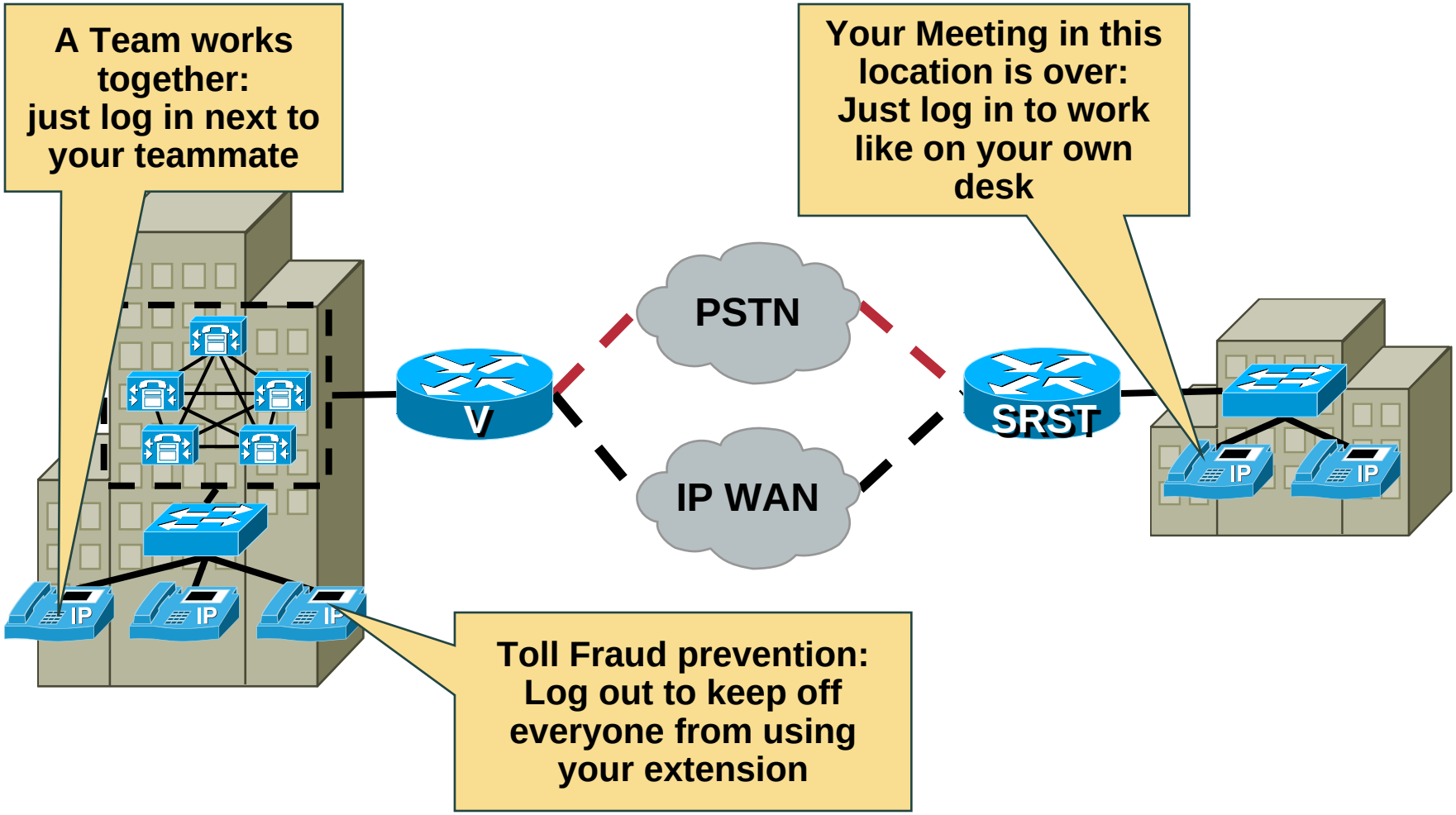
Extension Mobility Usage Examples



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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**

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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

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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**

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Attendant Console Overview
Client GUI Summary

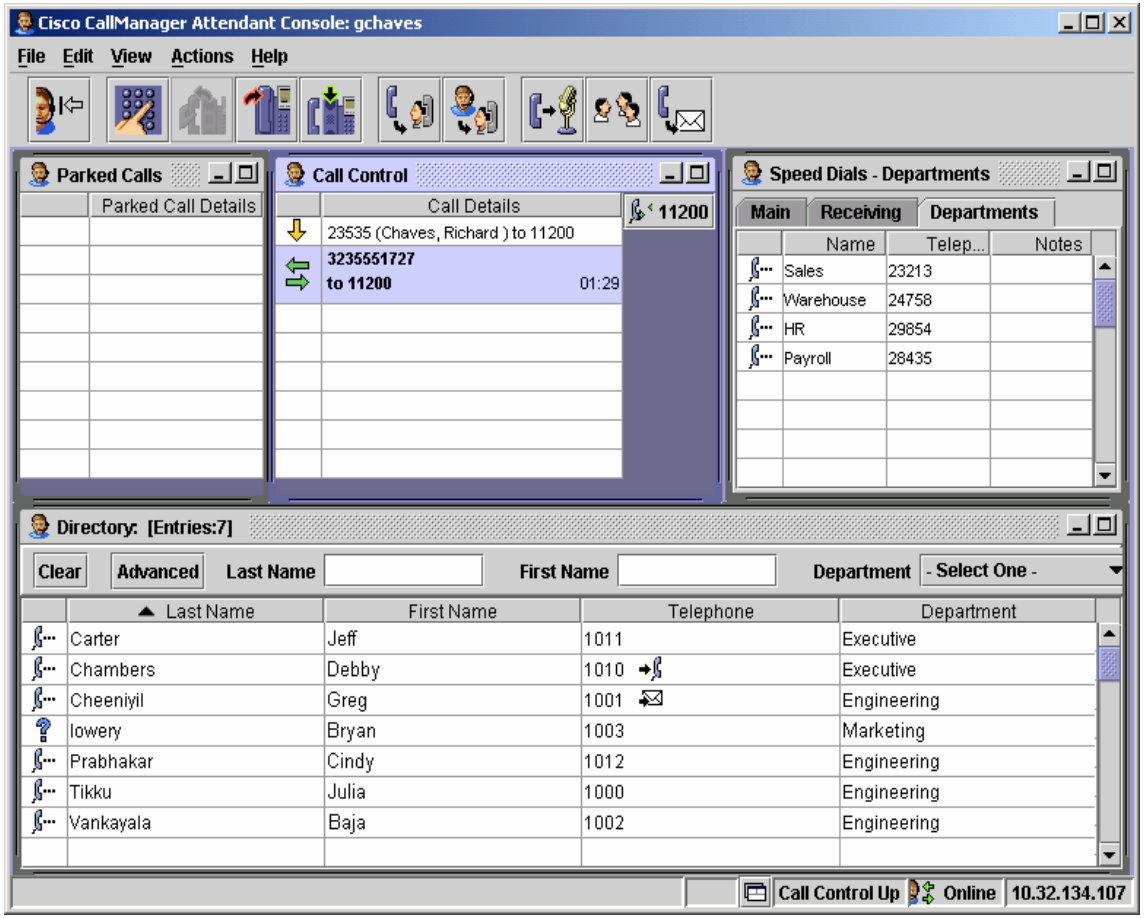
Drag-and-drop hold, transfer

Line states

- Idle line
- Inbound call
- Hold
- Active
- Unknown

Directory Pane

- Search by any field in the directory
- Sortable columns
- Call Forward status icons



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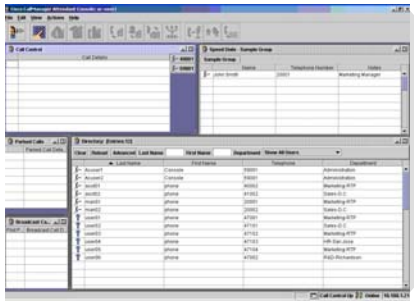
Attendant Console Overview: Hunt Groups



Attendant 2 answers call 4

Attendant 2

Line[1]: 60001
Line[2]: 59001

Attendant 1
Not logged in

Attendant 1

Line[1]: 40001
Line[2]: 59002

Pilot Point passes call to available member of Hunt Group 2

AC Pilot Point DN 10001 2

Hunt Group
1st Attendant 1, line 1
2nd Attendant 2, line 2

3 Call is passed to Attendant 2 DN: 59001 3

Dials: 10001

1 Phone A calls Pilot Point at 10001 1

Phone A

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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

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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**

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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

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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

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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.**

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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**

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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 stuff 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

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Cisco Unified CallManager 5.0 Features and Applications

Configuring Extension Mobility

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Objectives

- Extension Mobility Configuration Steps
- Extension Mobility Service Configuration
- Extension Mobility Configuration on IP Phone
- Create Profiles
- Configure User for Extension Mobility

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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

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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.**

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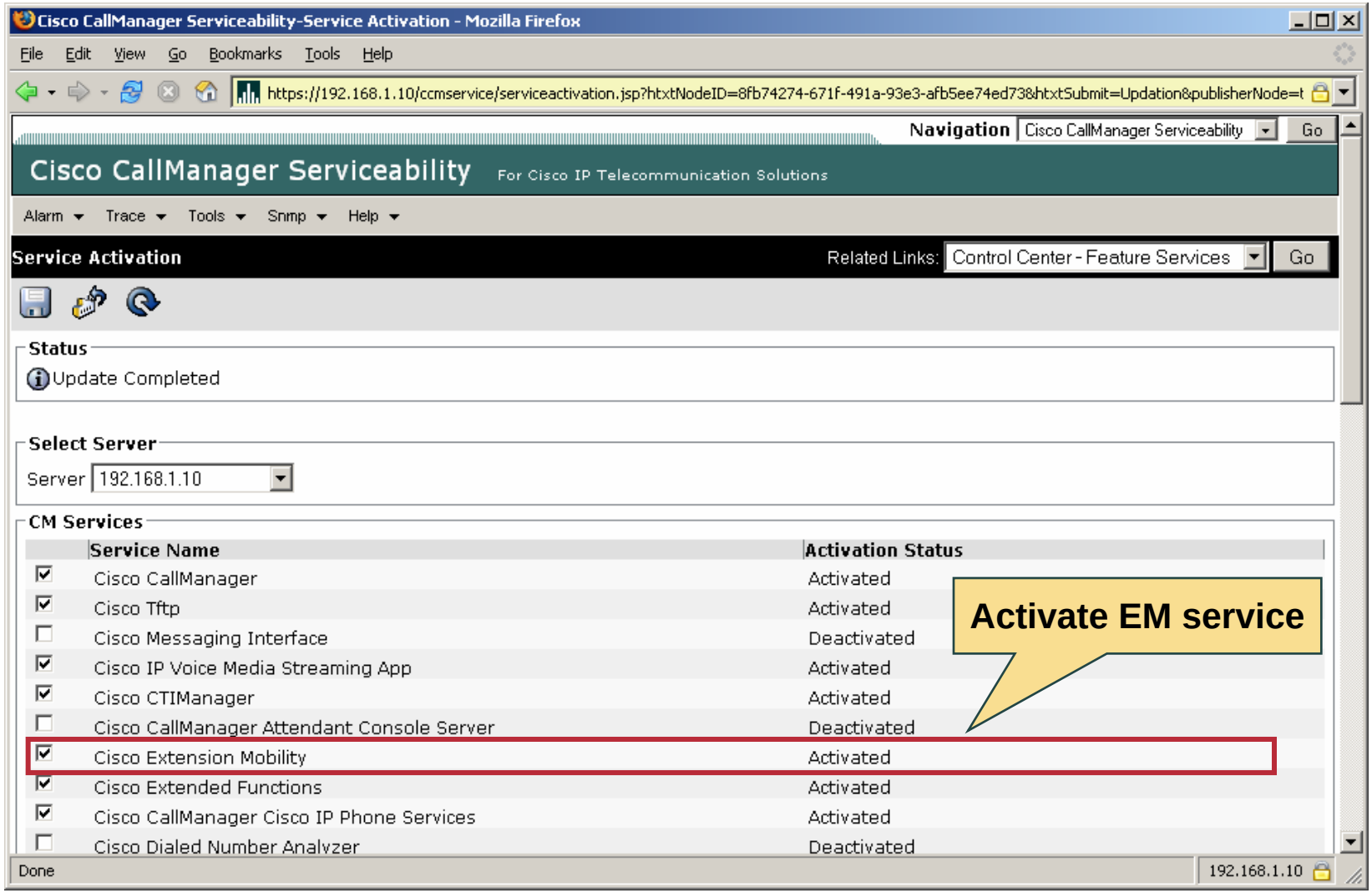
Extension Mobility Service Configuration: Enable Service



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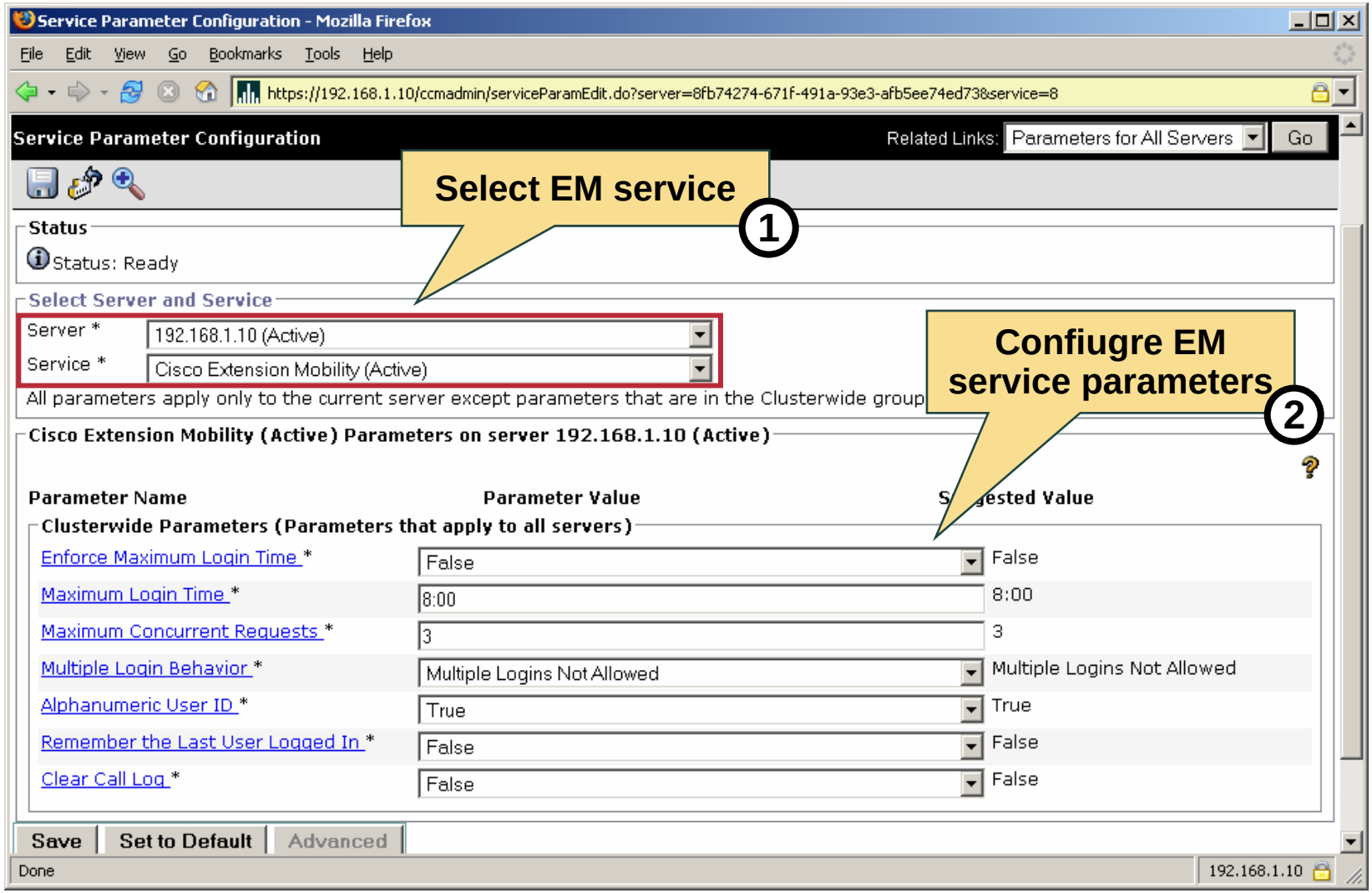
Extension Mobility Service Configuration: Service Parameters



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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 legged in on EM service logon.
Clear call log	False, True	Define if missed, placed, and received call directories are stored if logging on/off.

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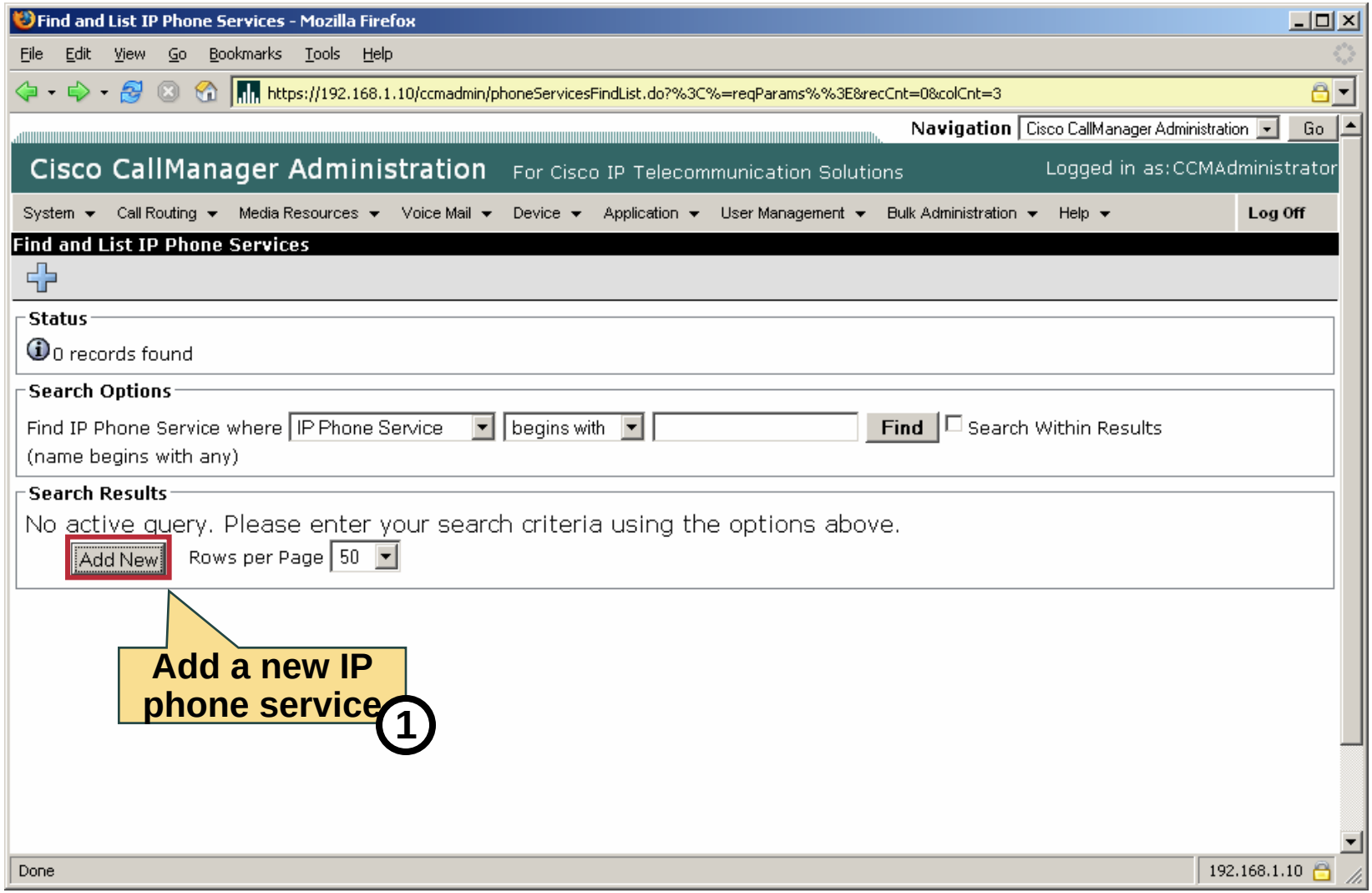
Extension Mobility Service Configuration: IP Phone Service



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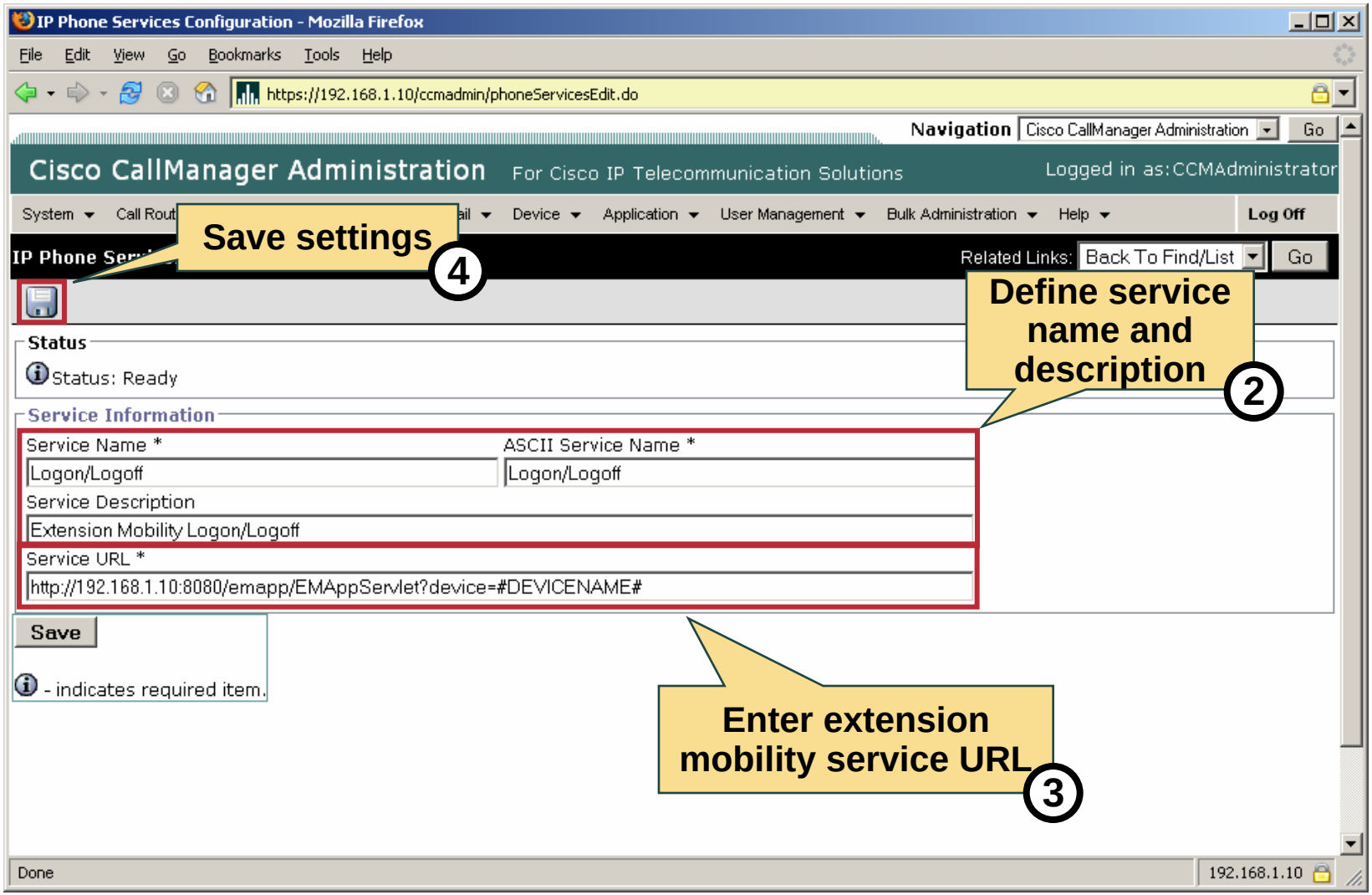
Extension Mobility Service Configuration: IP Phone Service



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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.**

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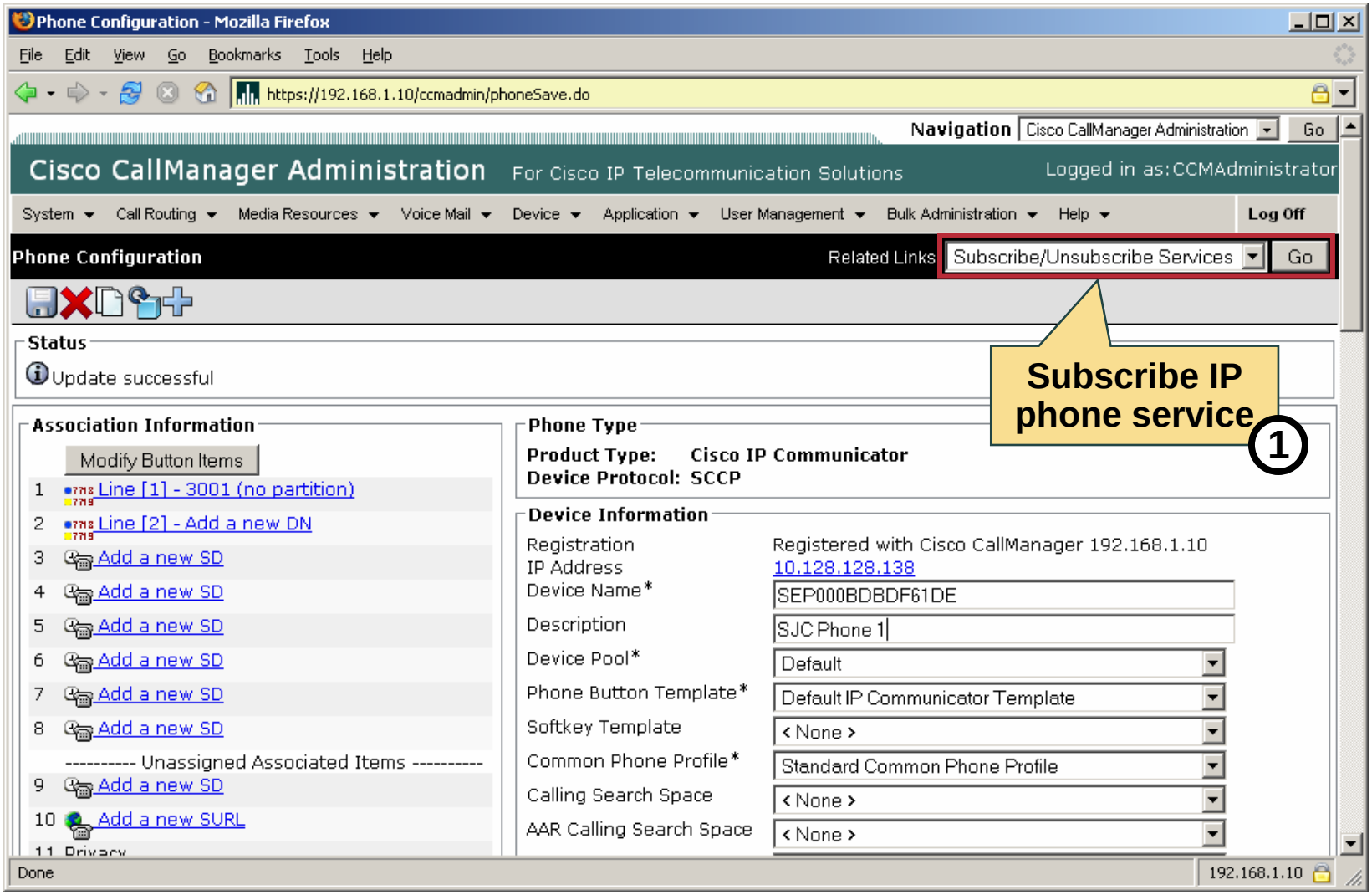
Extension Mobility Configuration on IP Phone



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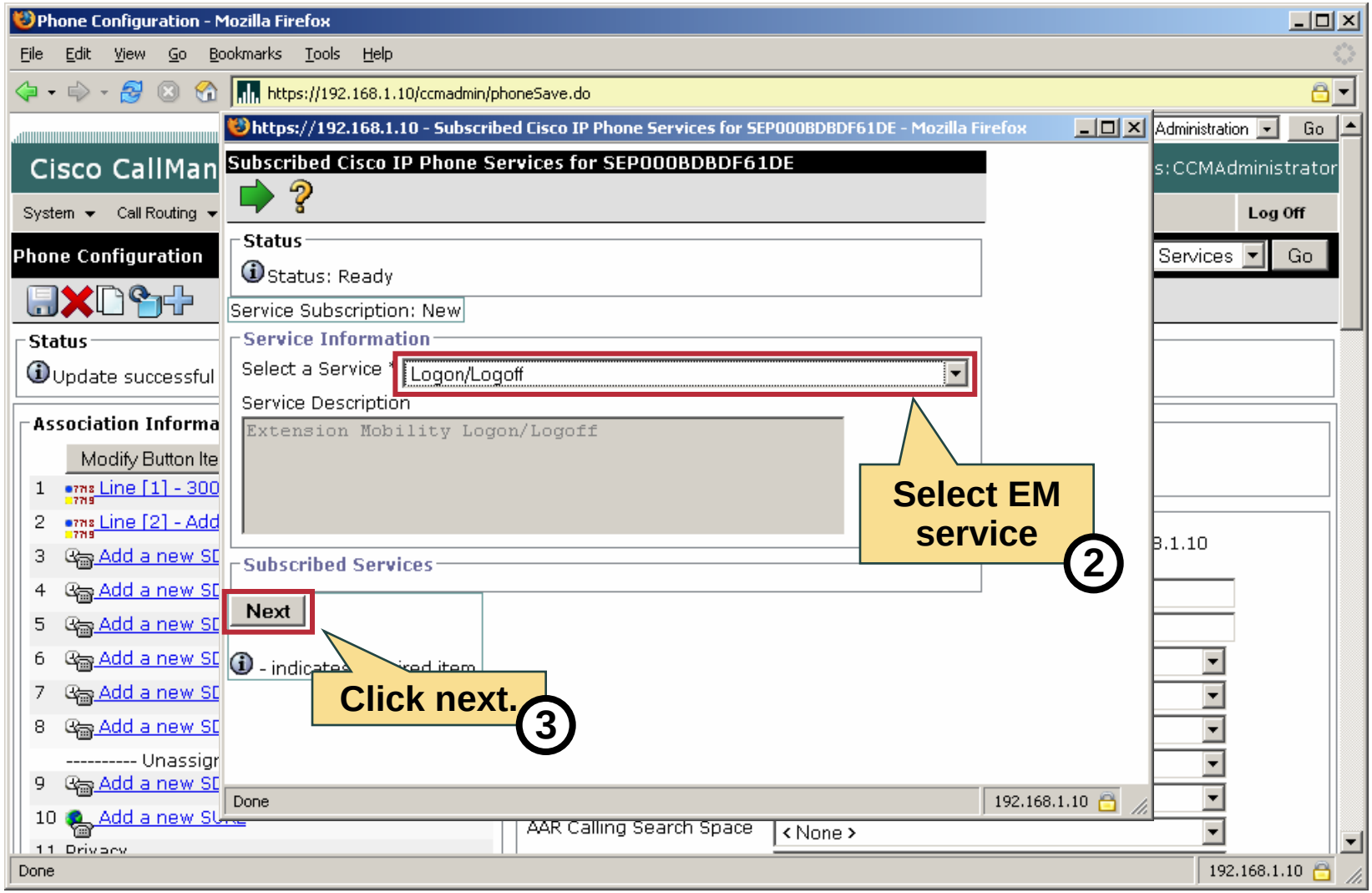
Extension Mobility Configuration on IP Phone



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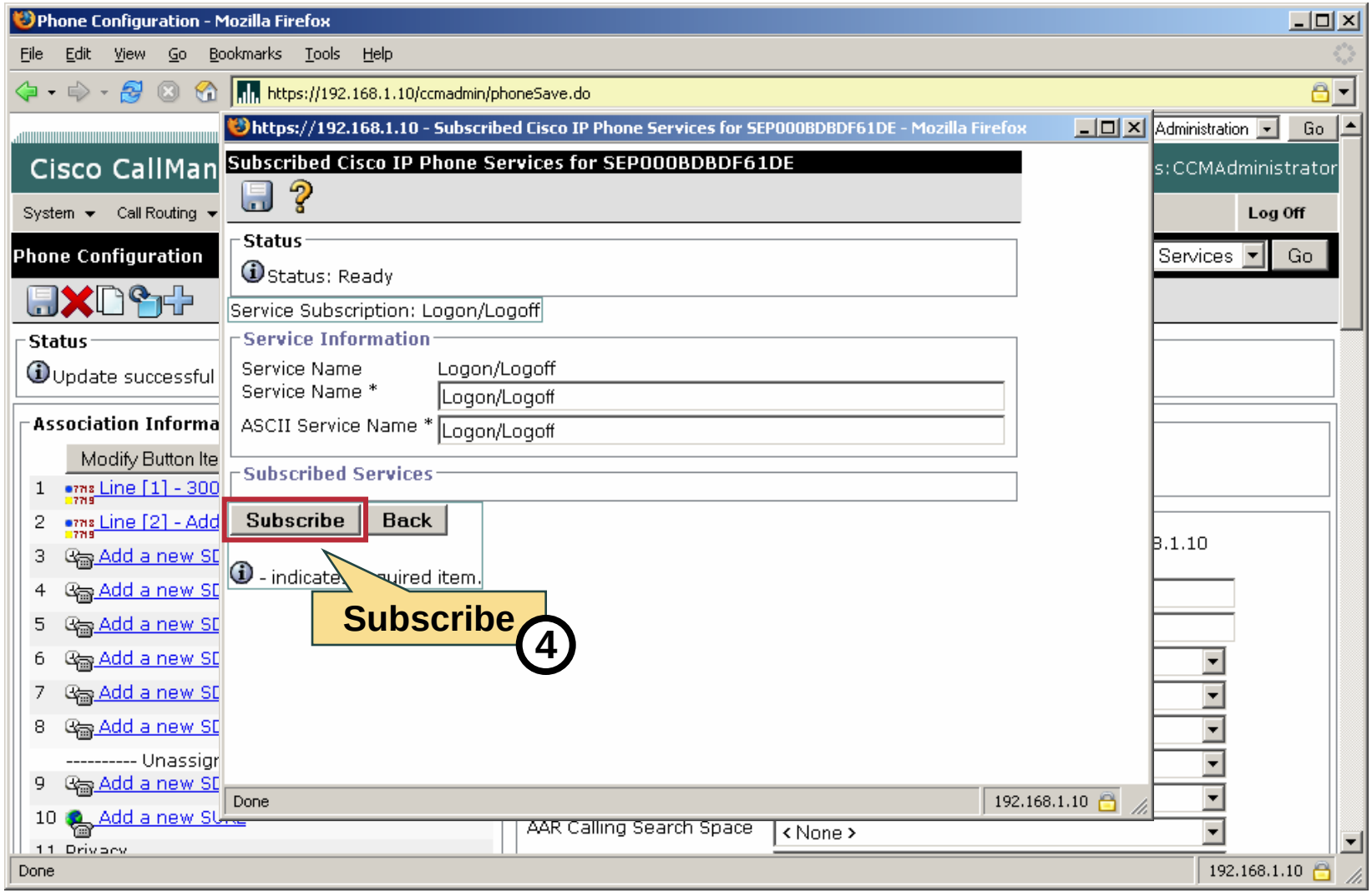
Extension Mobility Configuration on IP Phone



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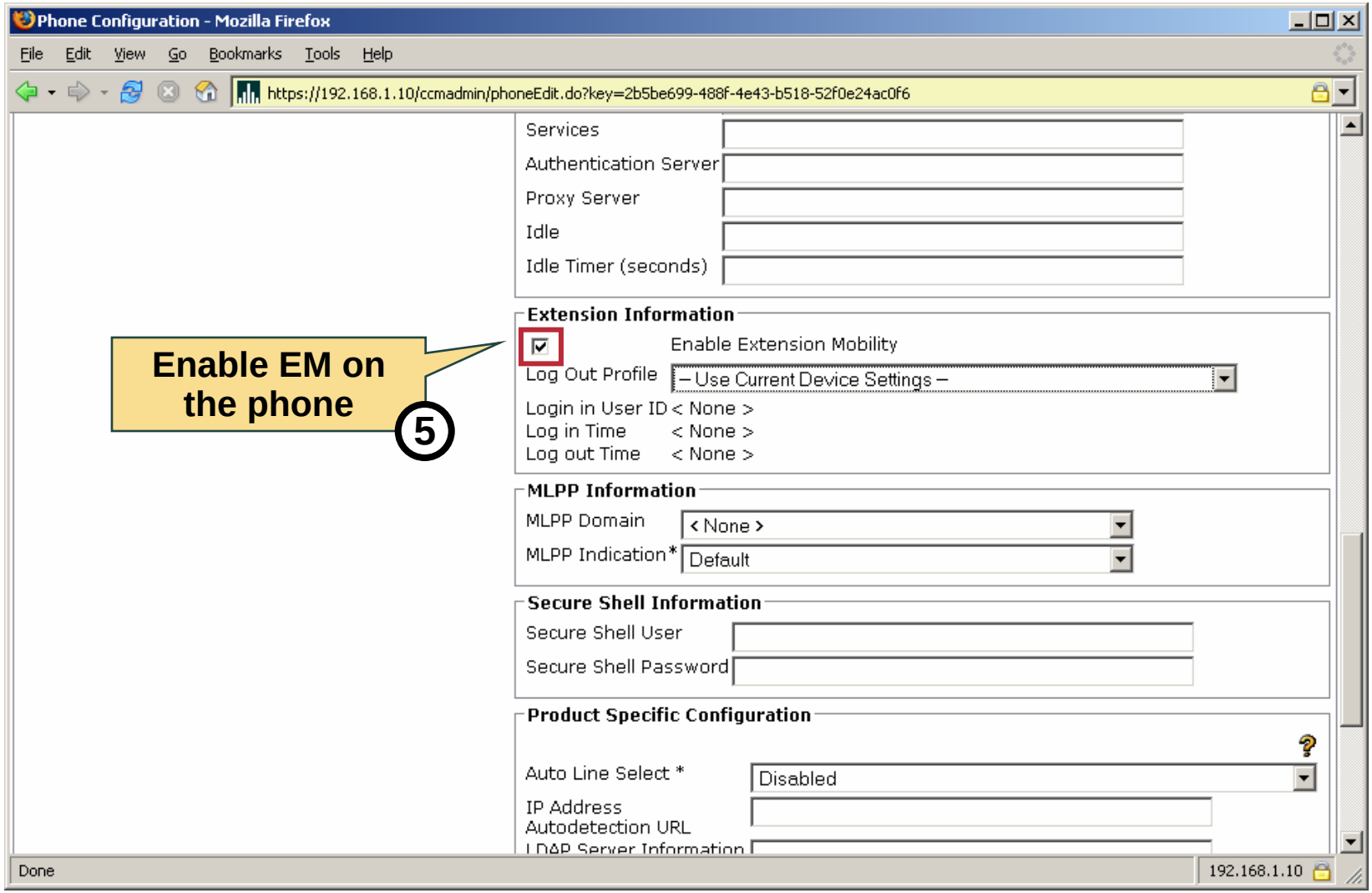
Extension Mobility Configuration on IP Phone



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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.**

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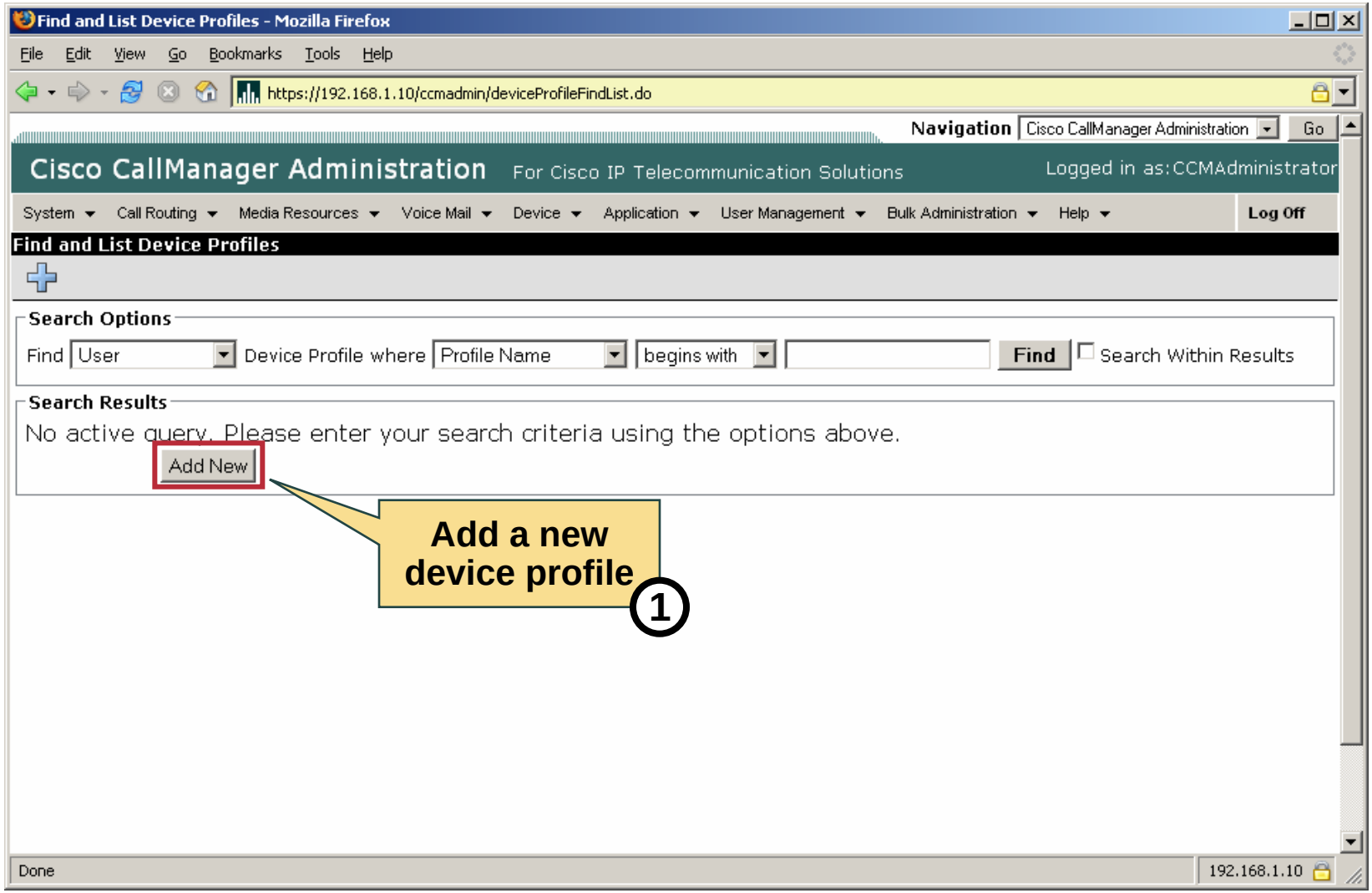
Create Profiles



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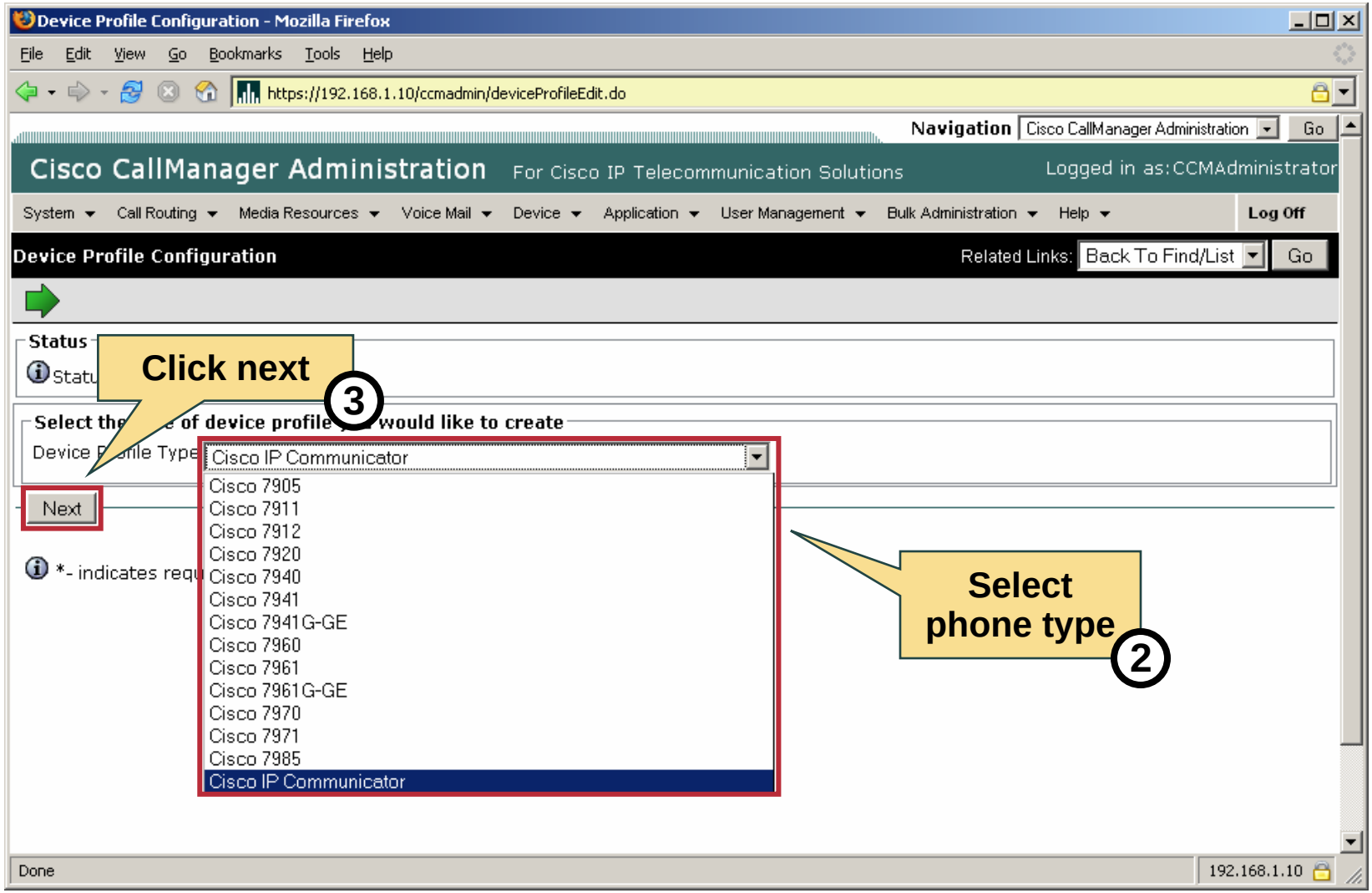
Create Profiles



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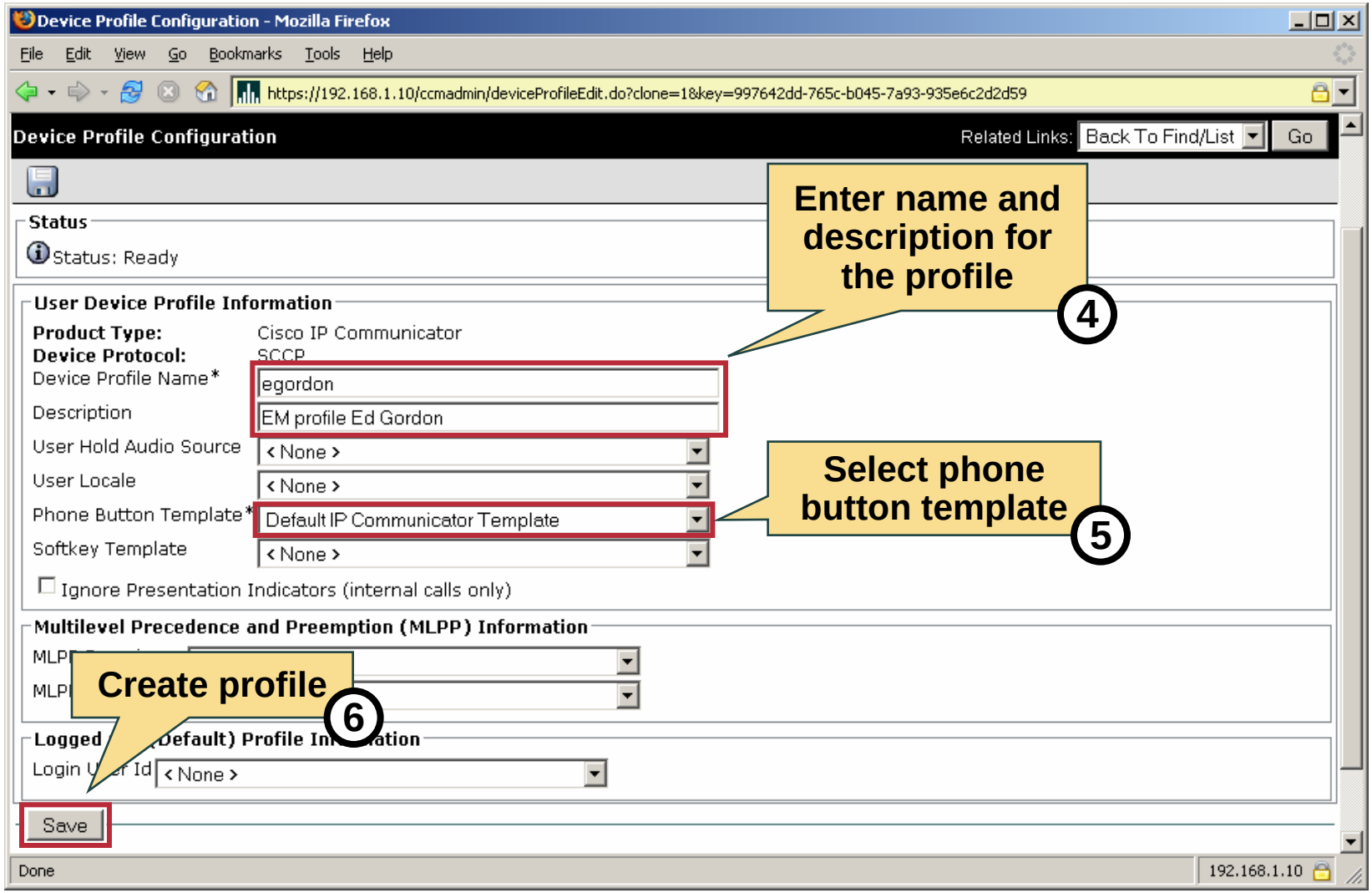
Create Profiles



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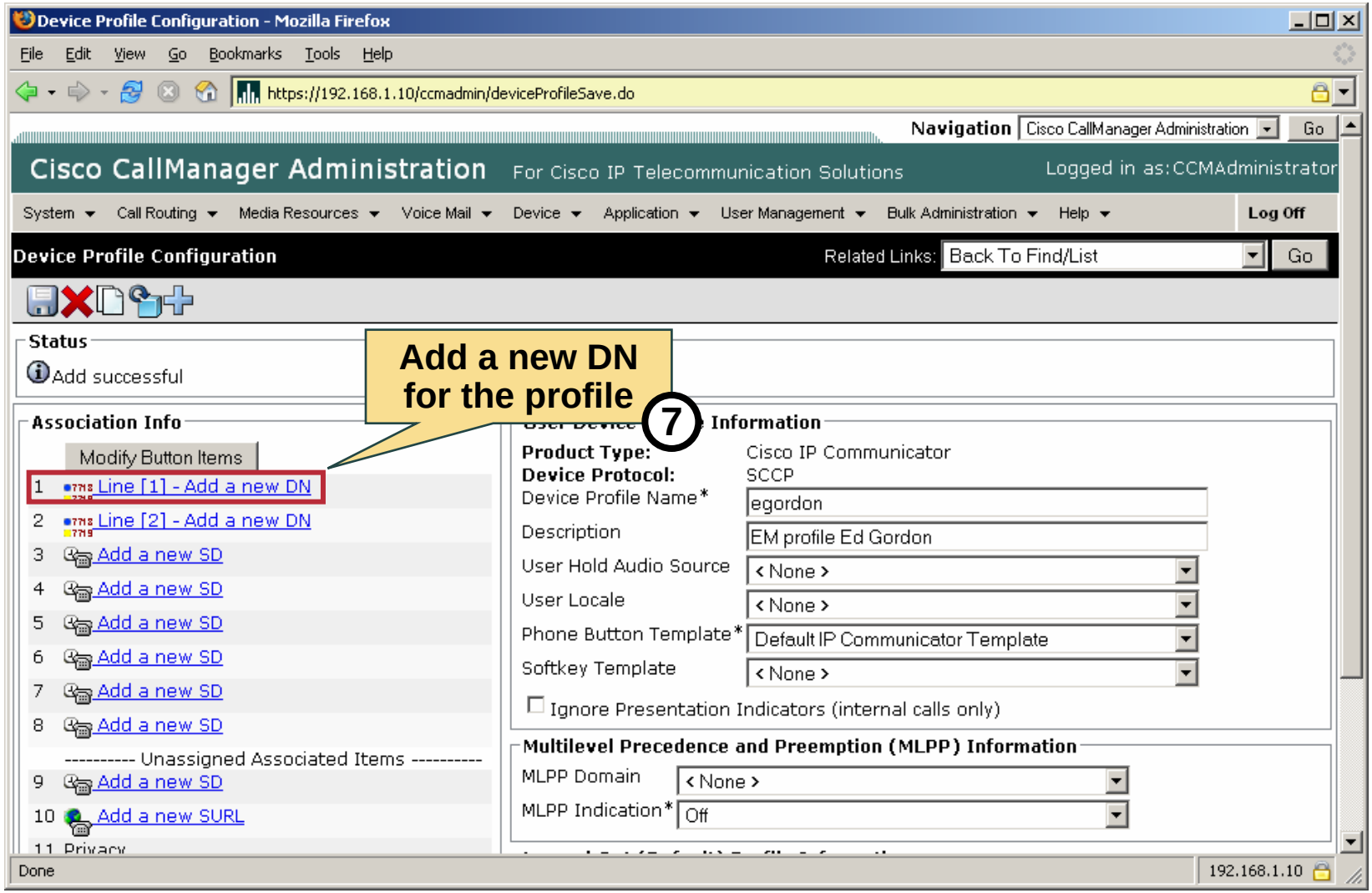
Create Profiles



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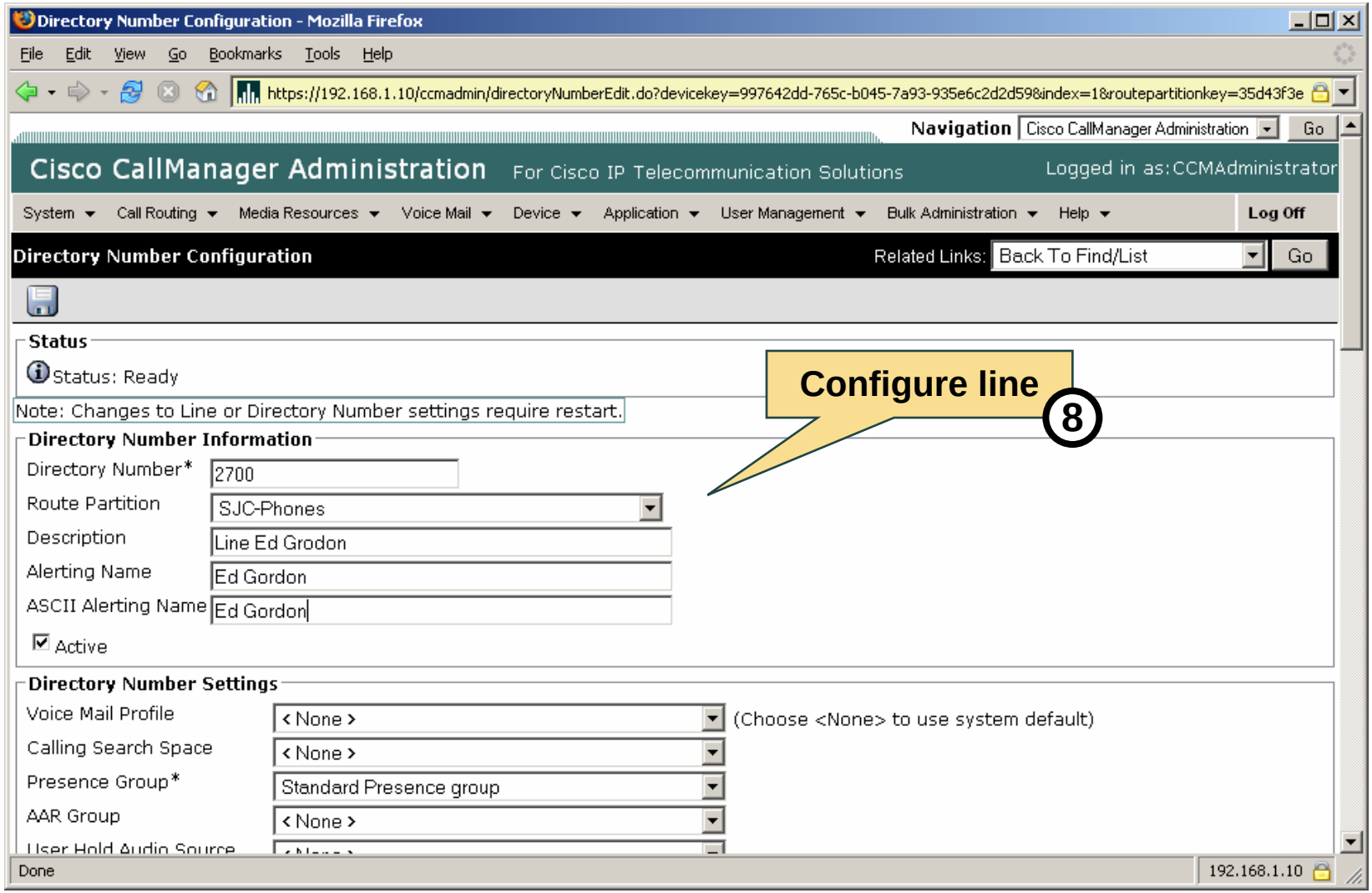
Create Profiles



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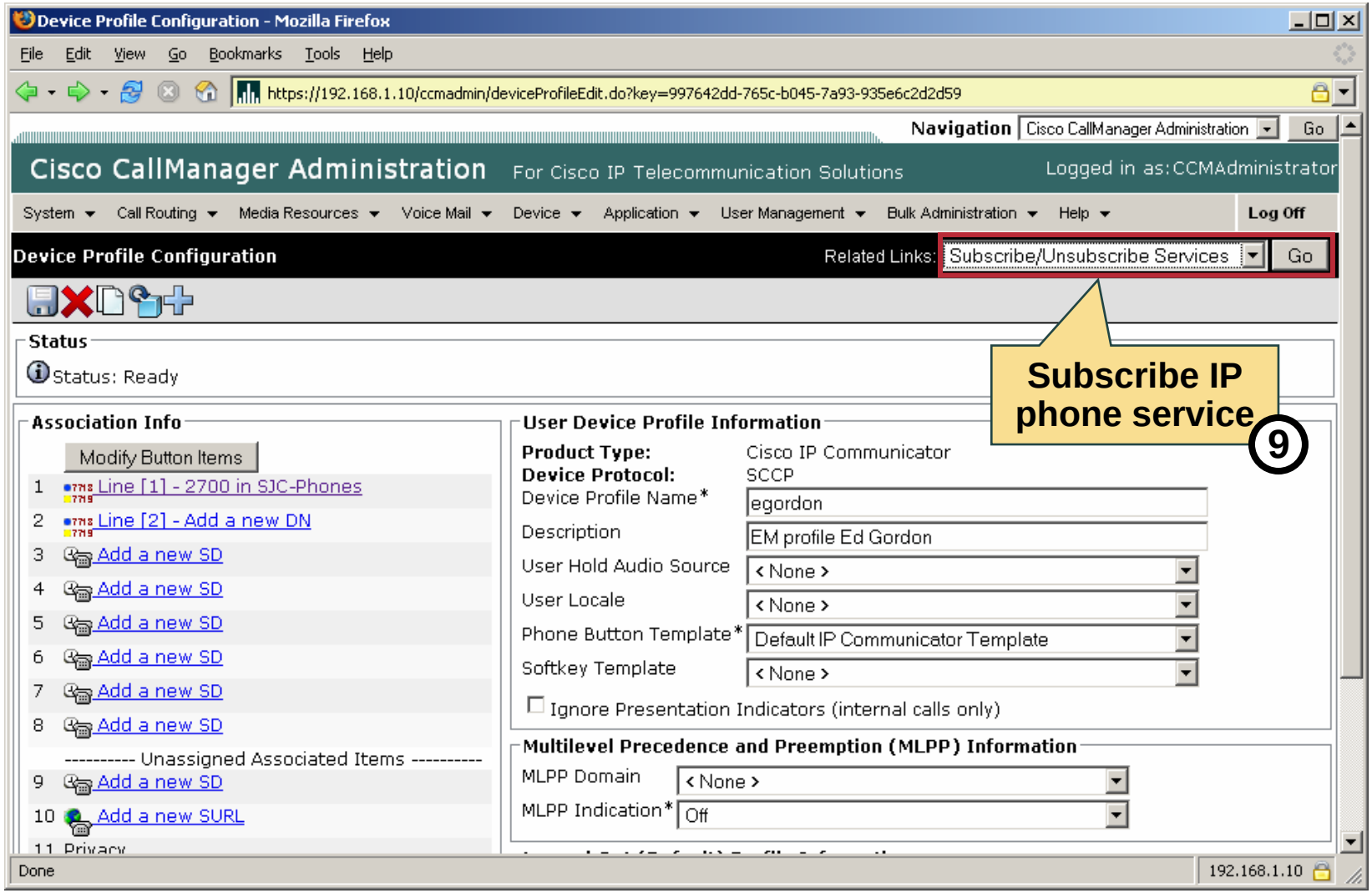
Create Profiles



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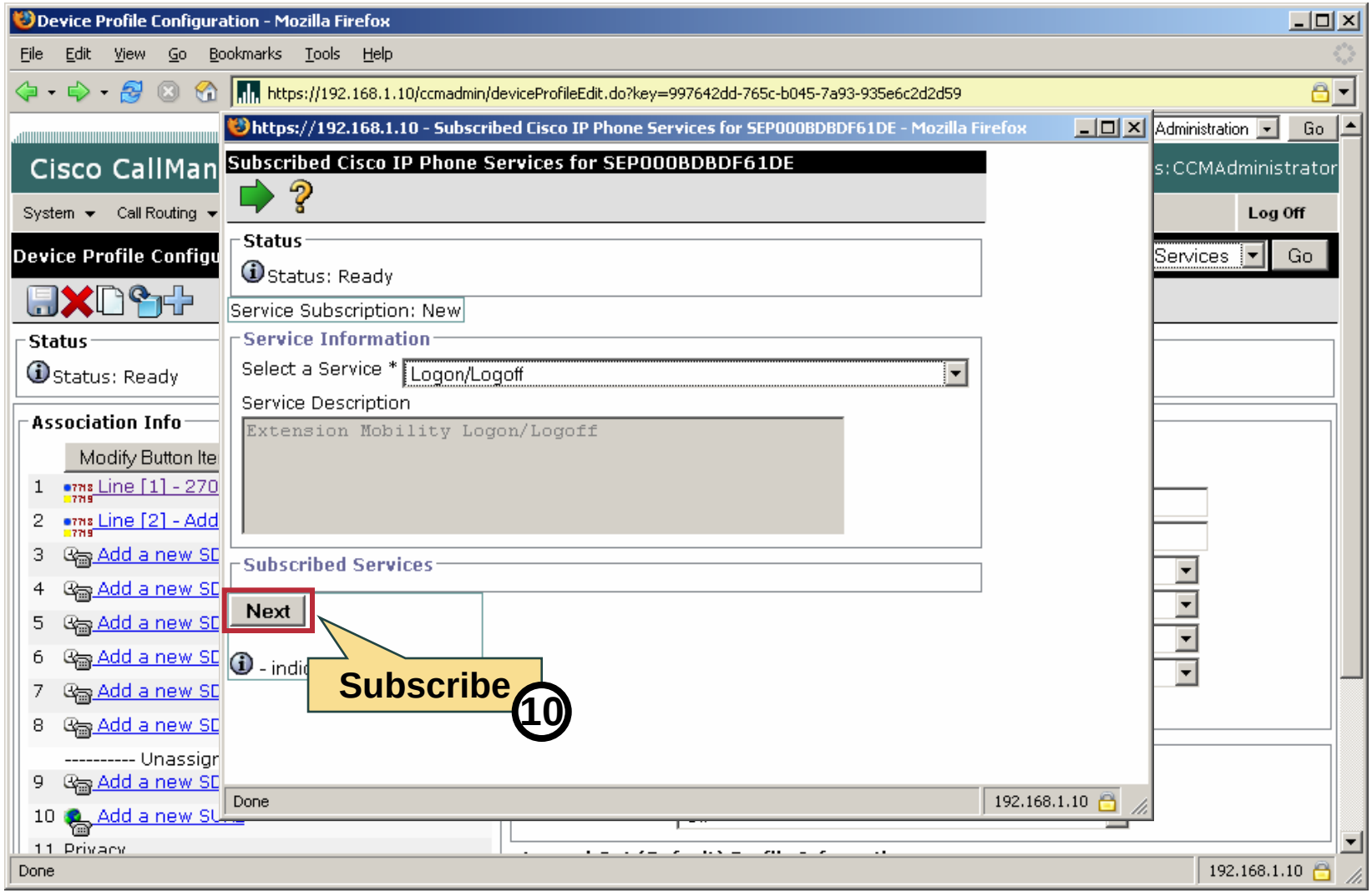
Create Profiles



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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.**

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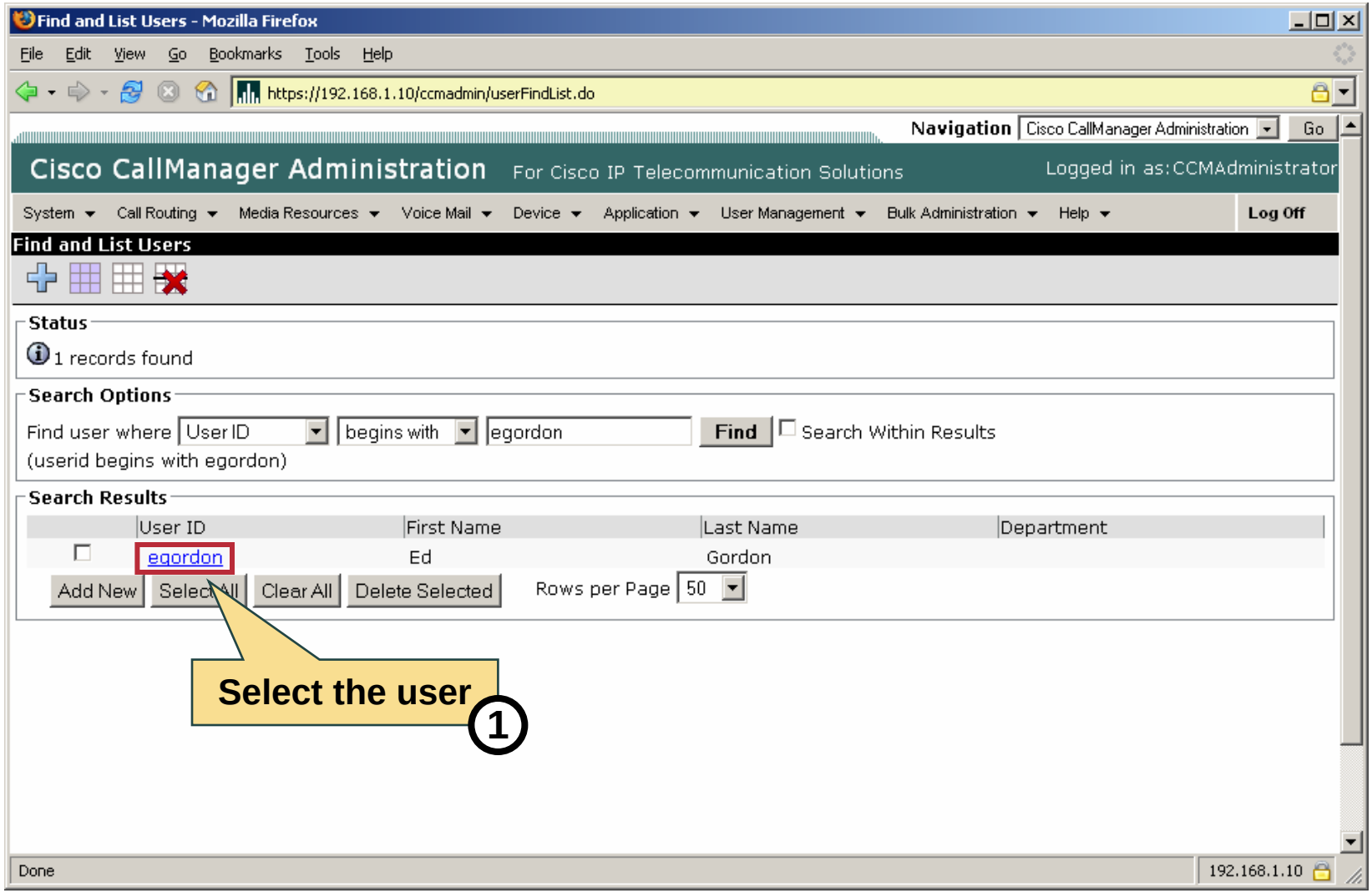
User Configuration



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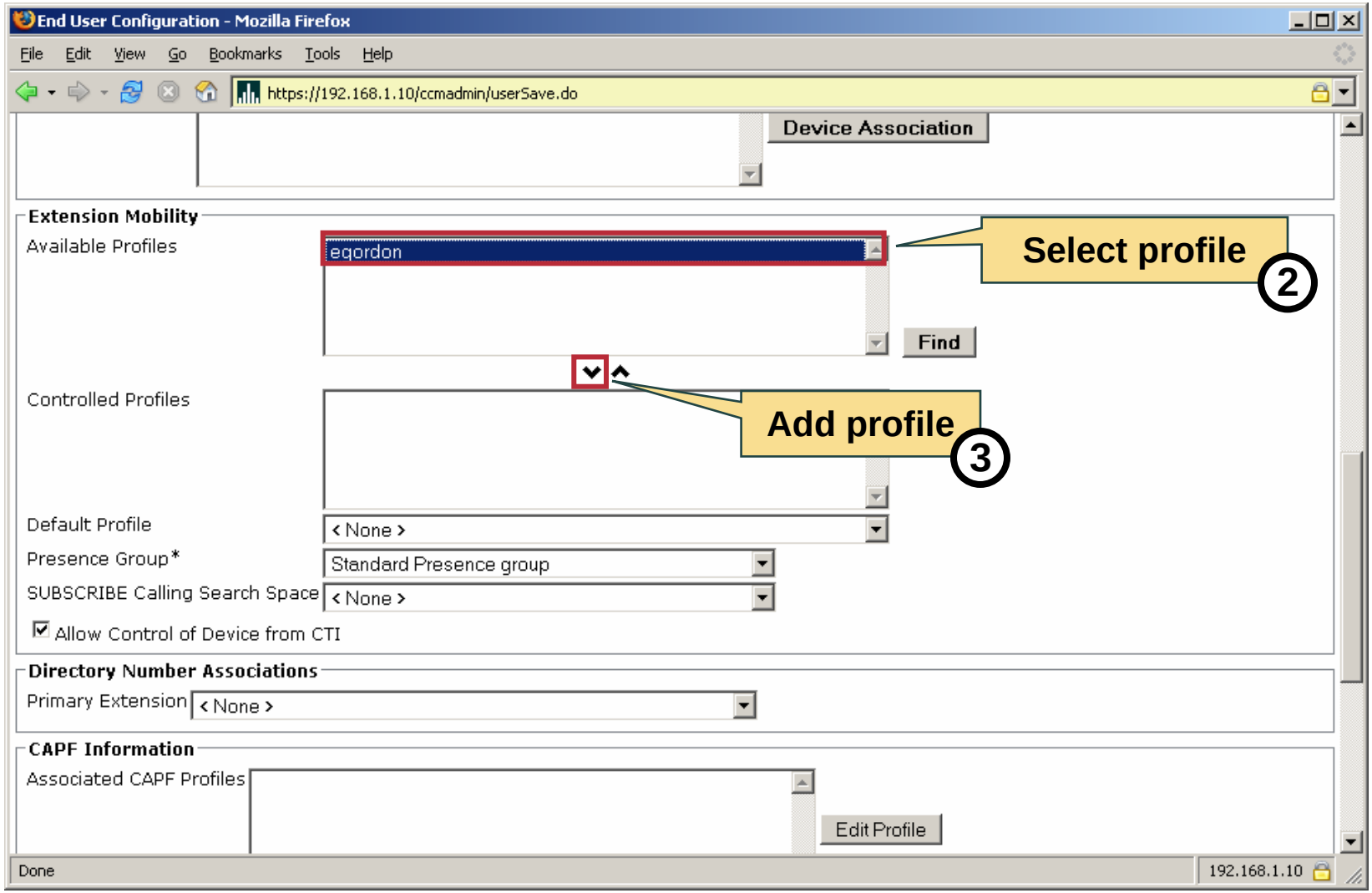
User Configuration



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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.

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Cisco Unified CallManager 5.0 Features and Applications

Presence and Busy Lamp Field

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Objectives

- **Speed Dial BLF and Call List BLF**
- **Configure BLF**
- **Configure Phones for BLF**
- **Configure Presence Authorization**

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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.**

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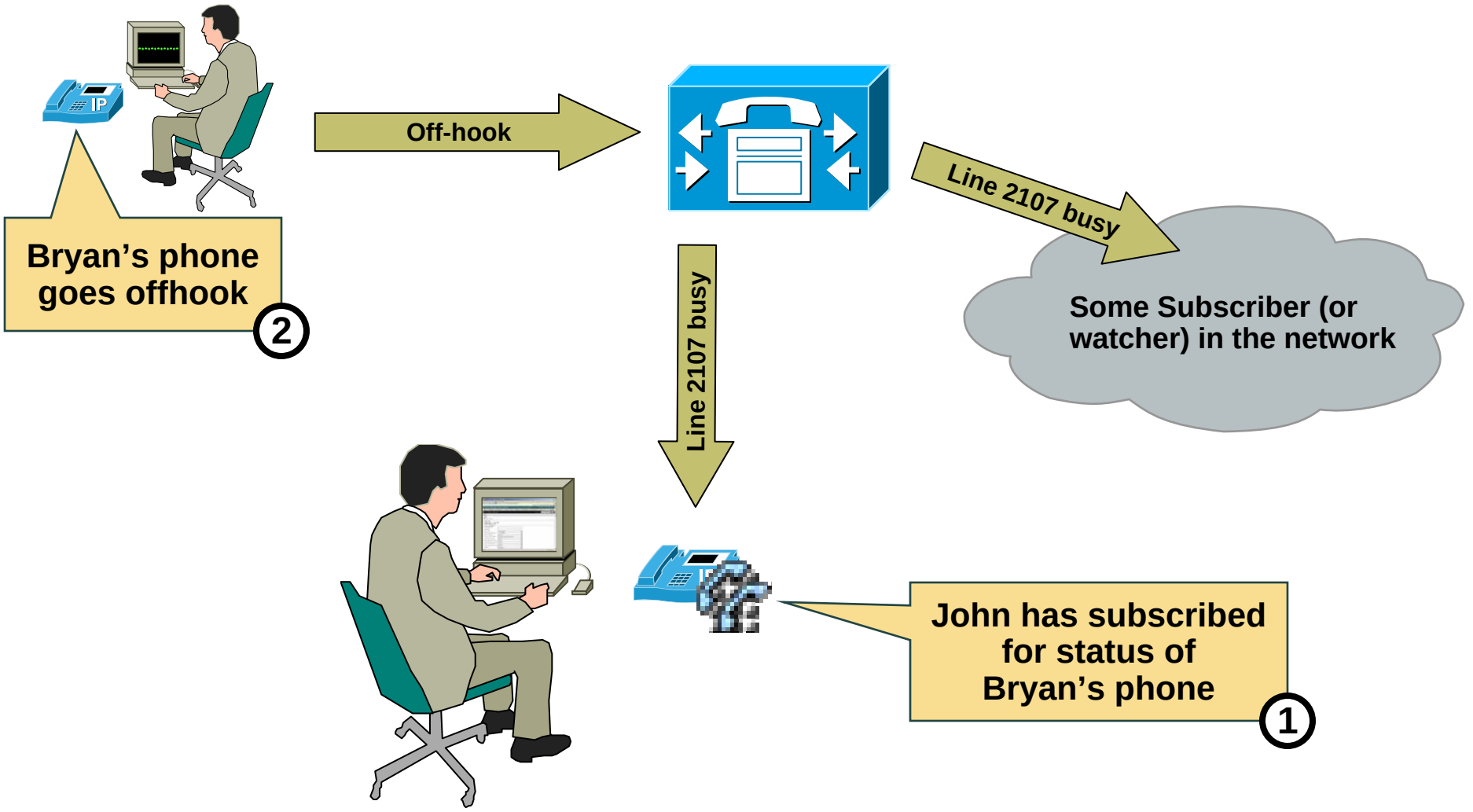
Speed Dial BLF and Call List BLF



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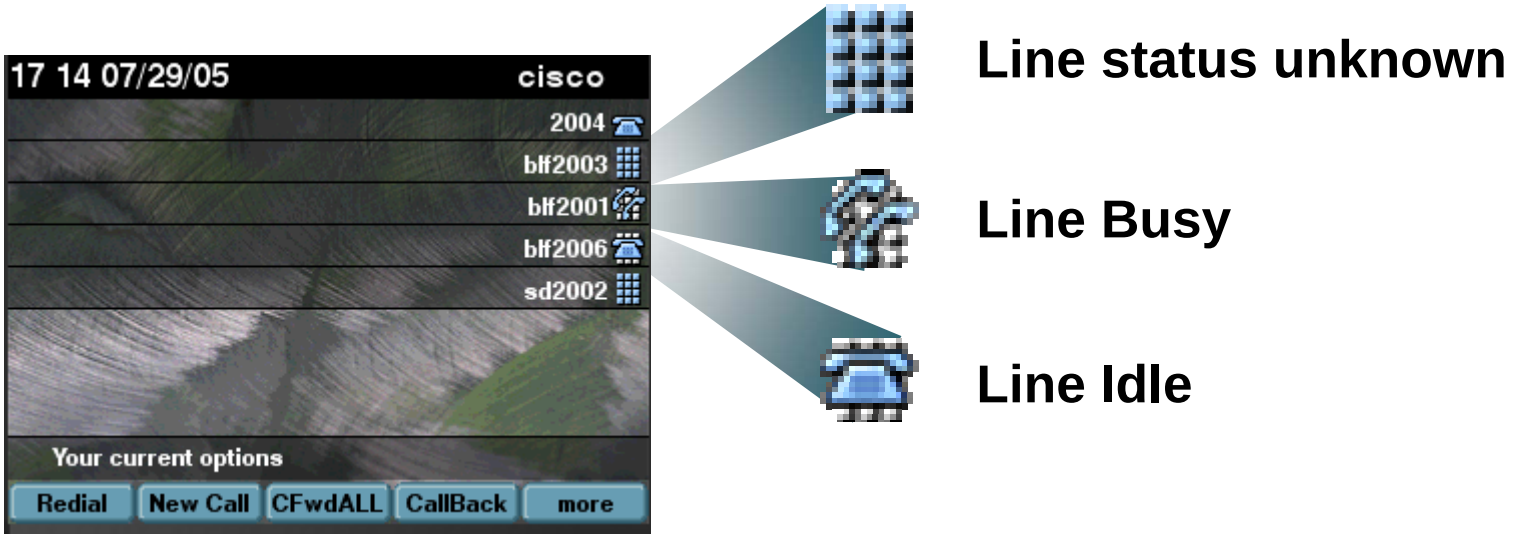
Speed Dial BLF and Call List BLF

Speed dial BLF:

- Permanently displaying the status of the speed dial DN or URL.
- Is supported on Cisco IP phones 7940 and greater.



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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.

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Speed Dial BLF and Call List BLF

Group based authorization

- Group based mechanism to control which subscribers (watchers) can see which subscribes (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)

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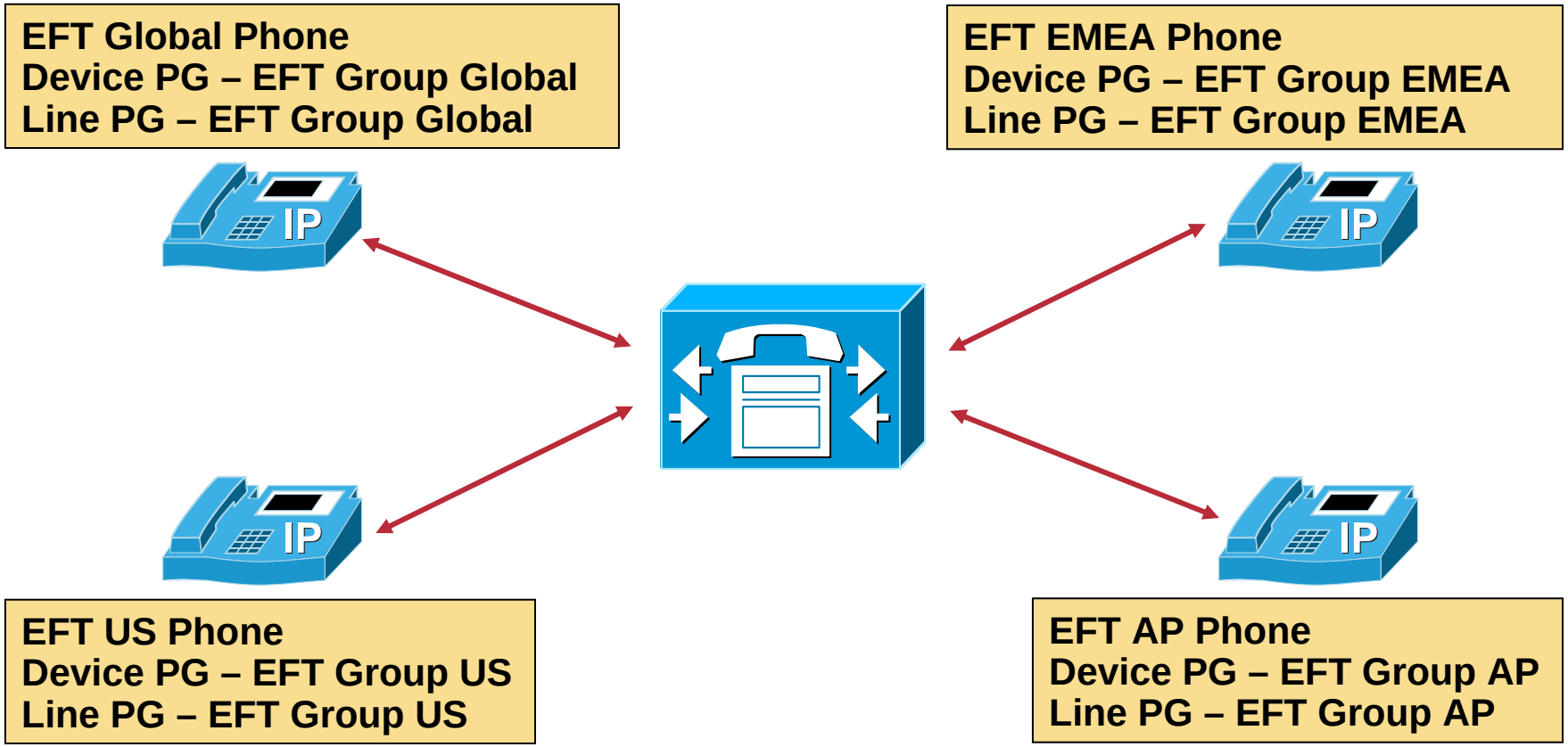
Speed Dial BLF and Call List BLF

Presence Group (PG) authorization example:



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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.**

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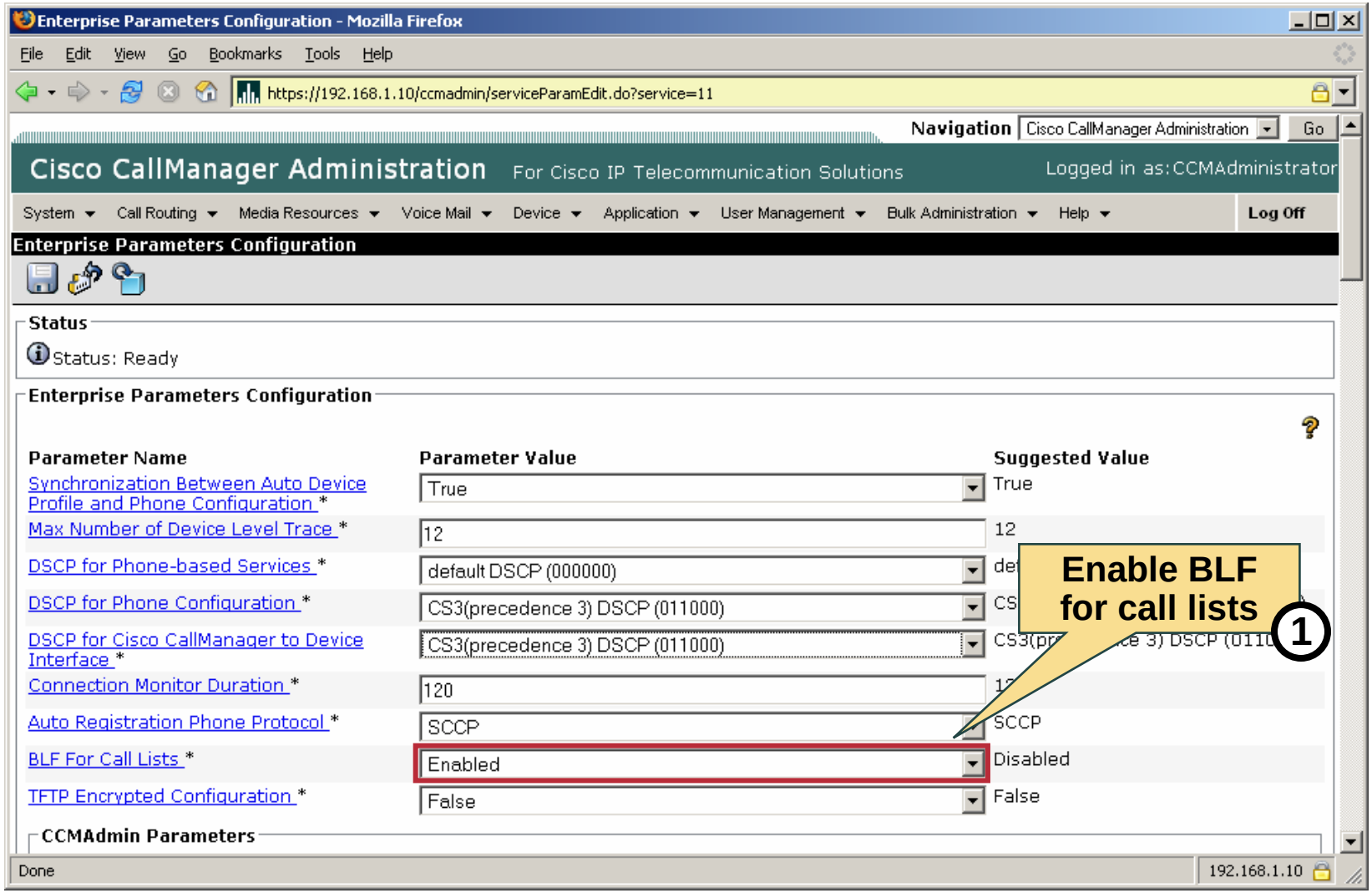
Configure BLF



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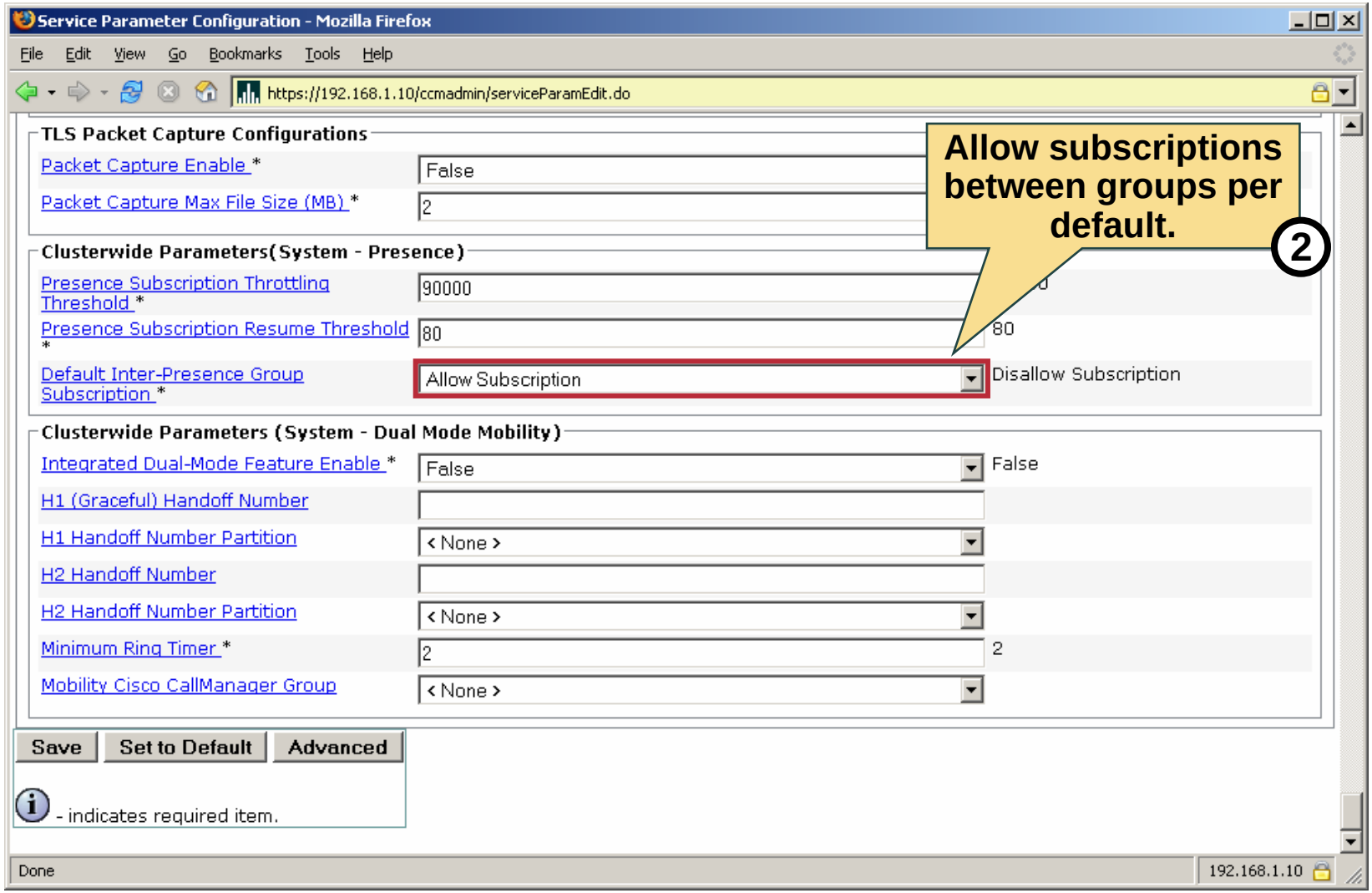
Configure BLF



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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.**

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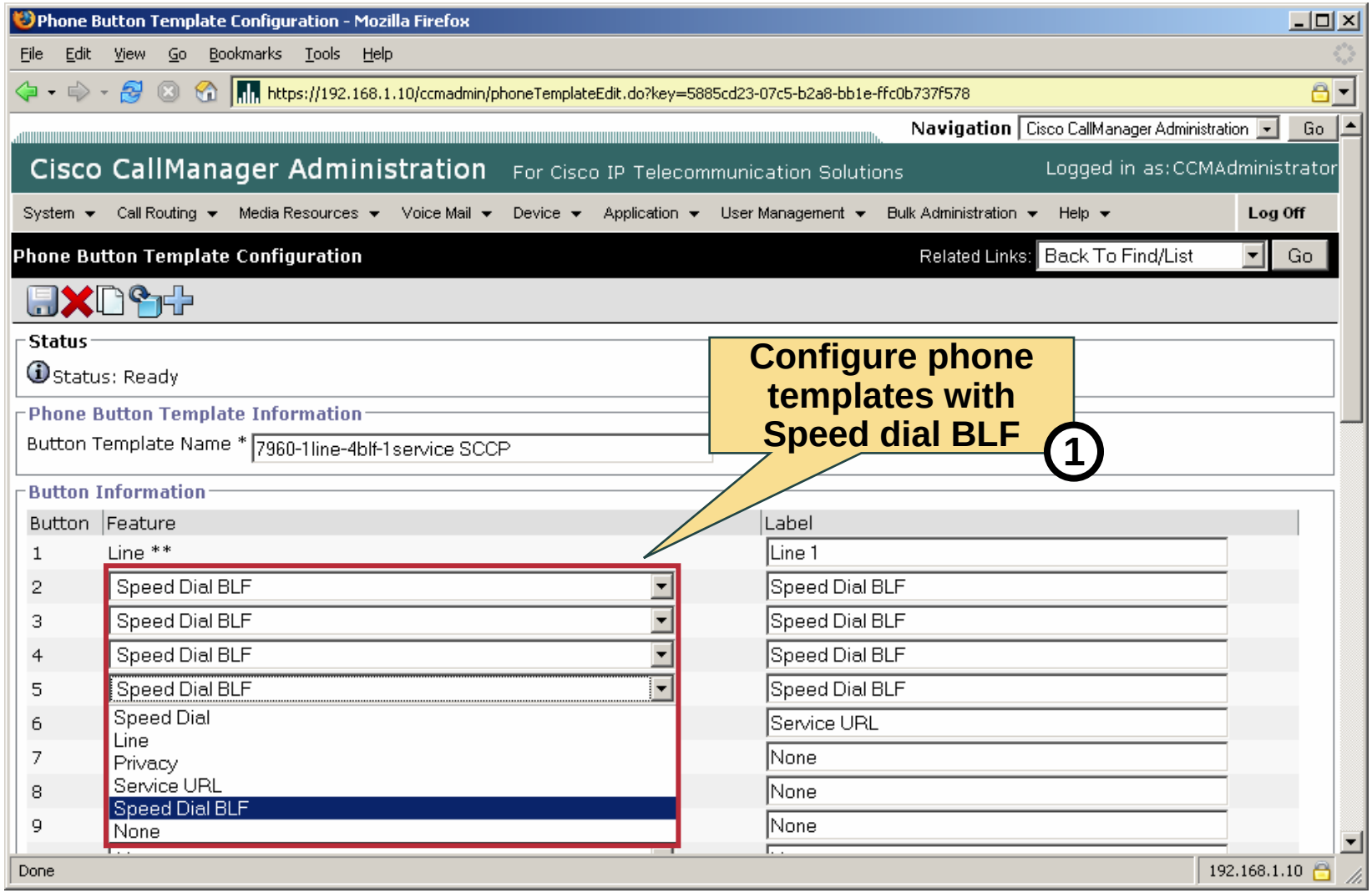
Configure Phones for BLF



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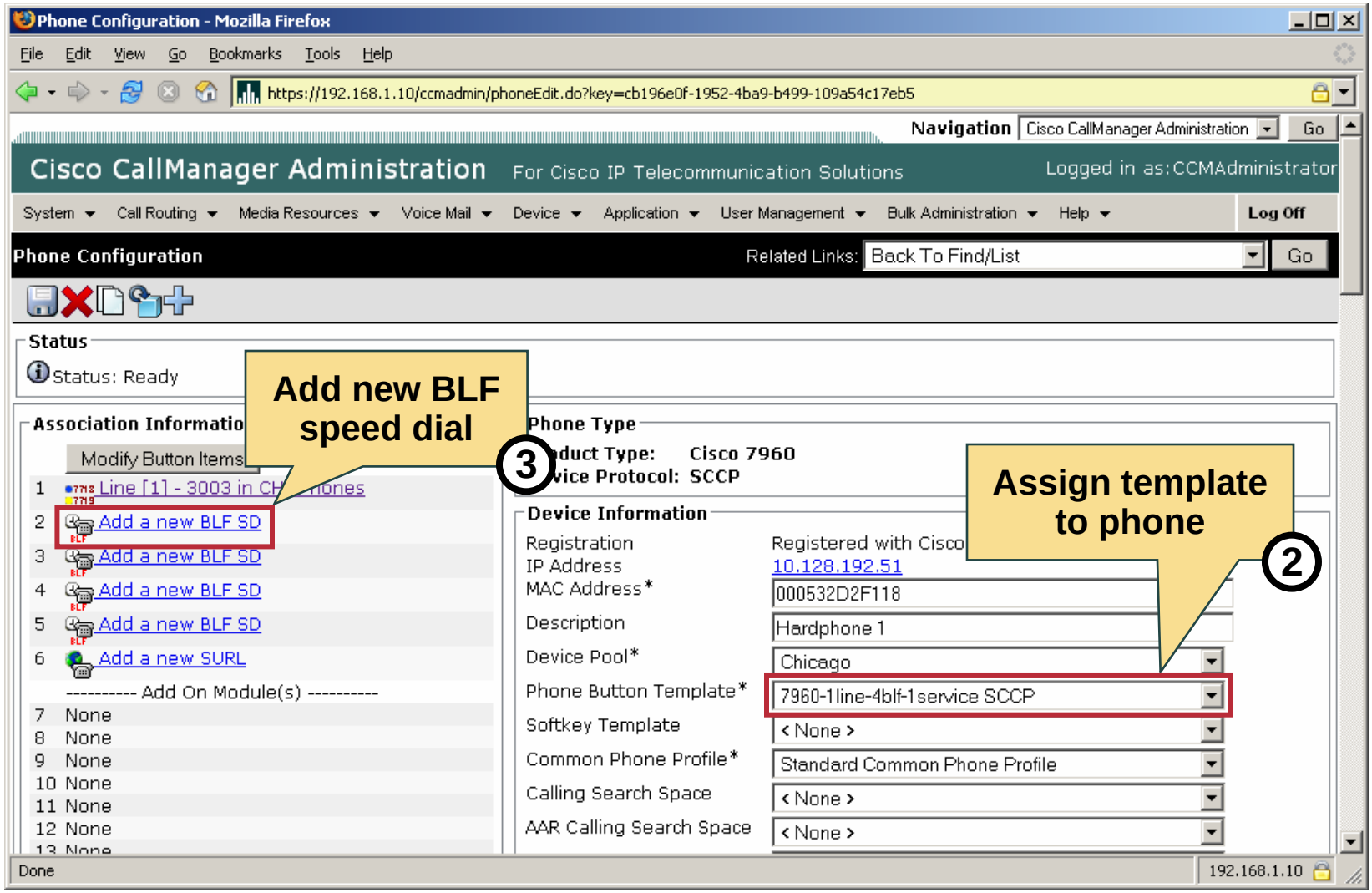
Configure Phones for BLF



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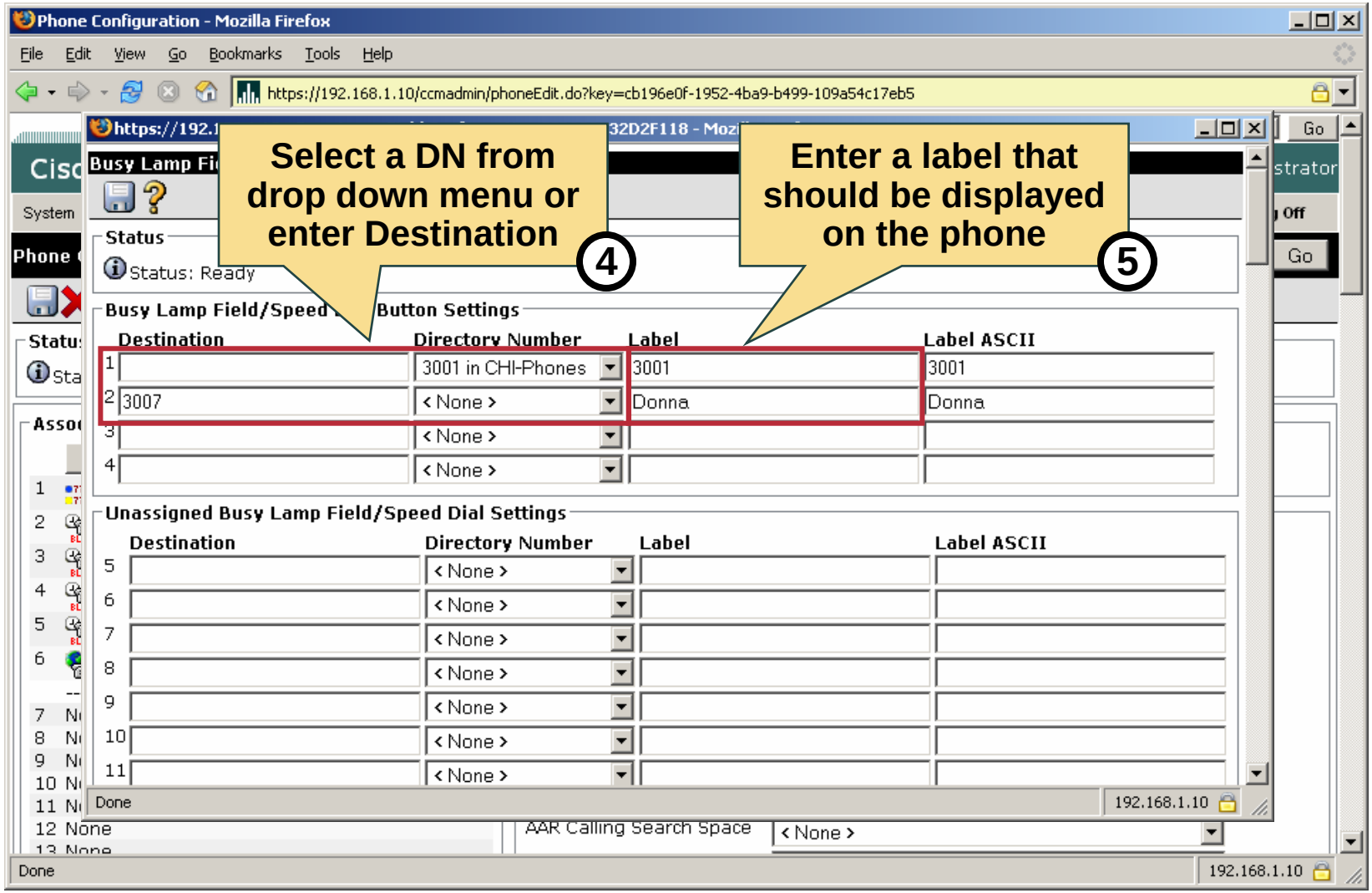
Configure Phones for BLF



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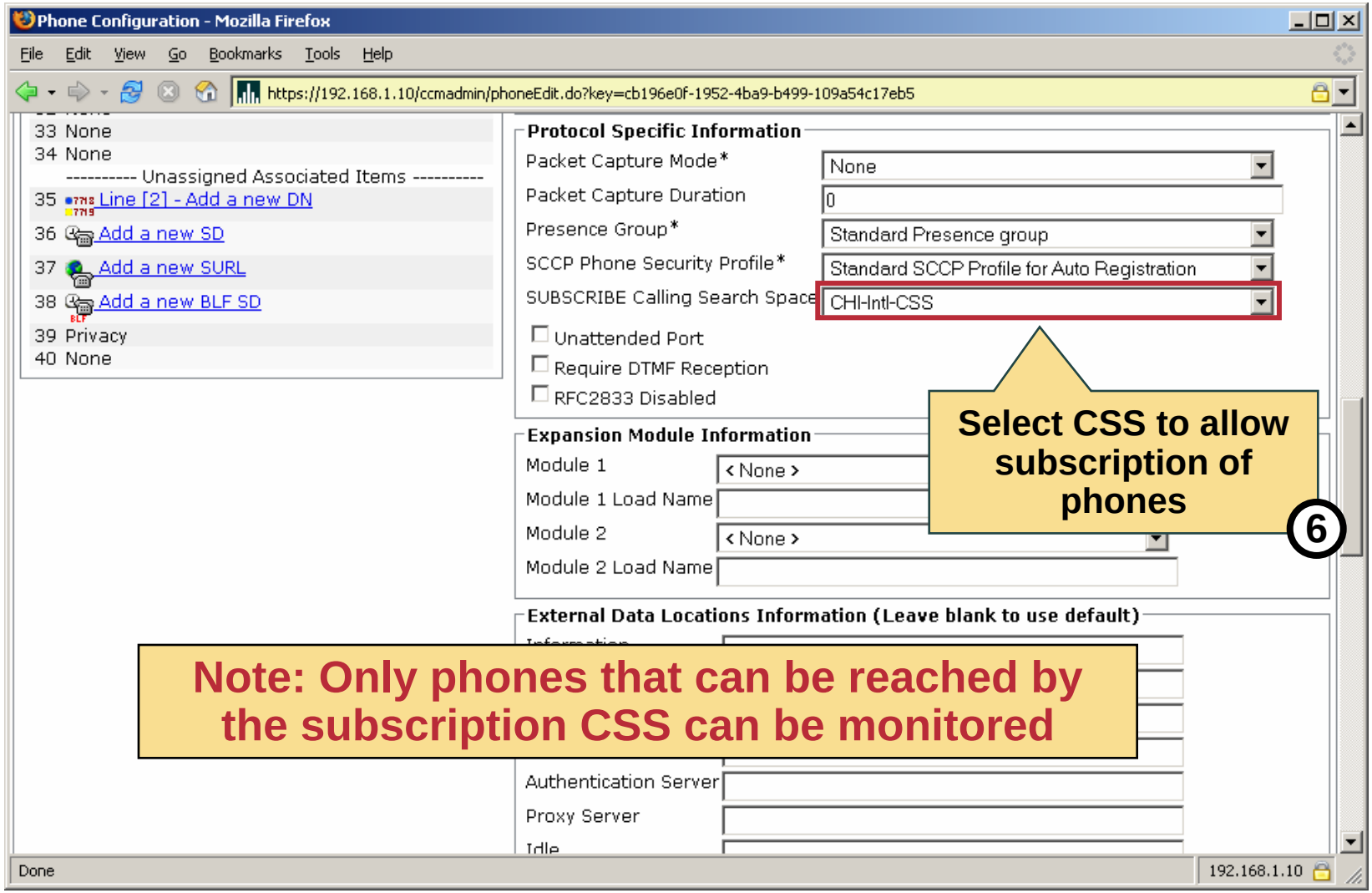
Configure Phones for BLF



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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.**

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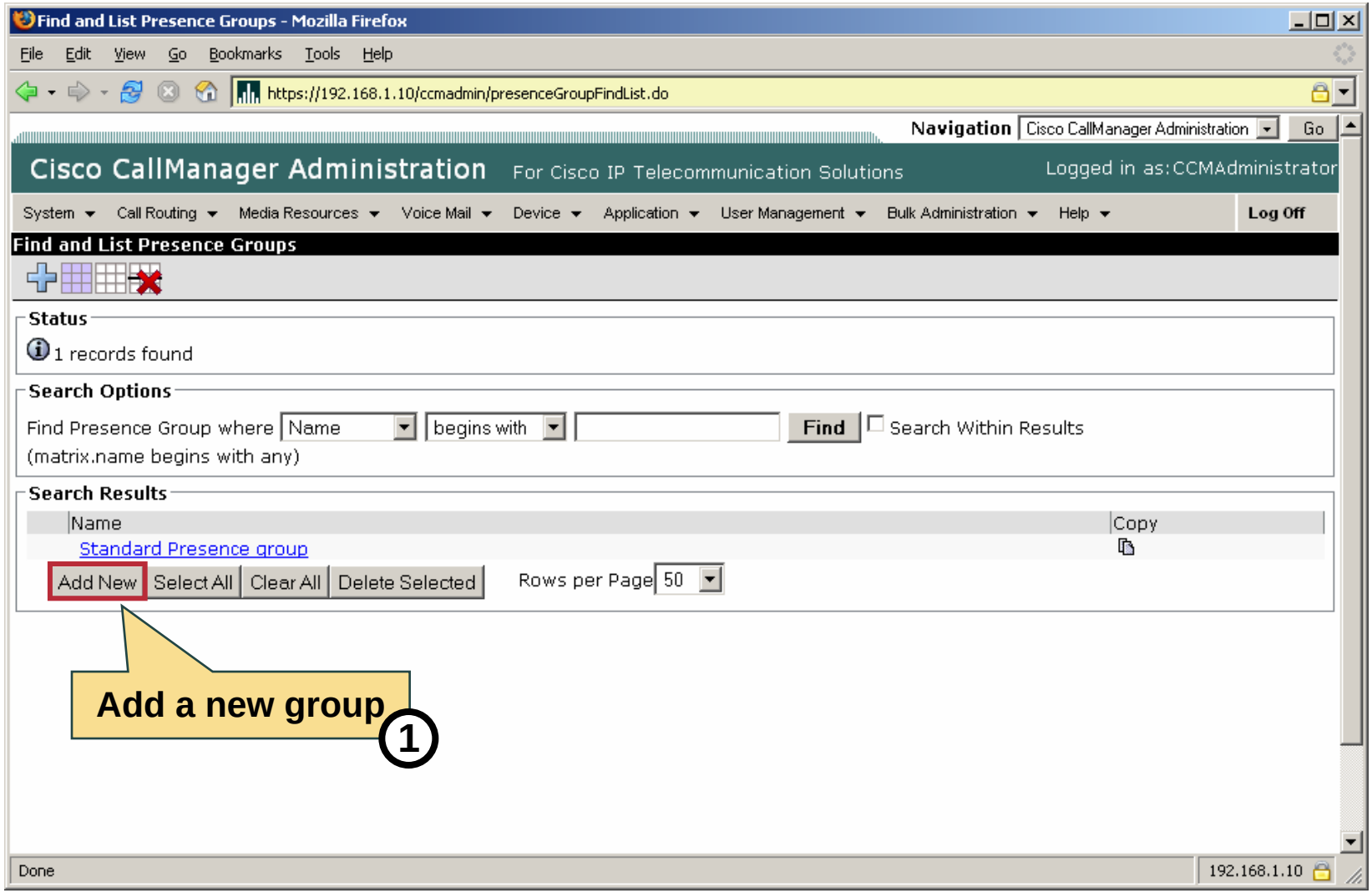
Configure Presence Authorization



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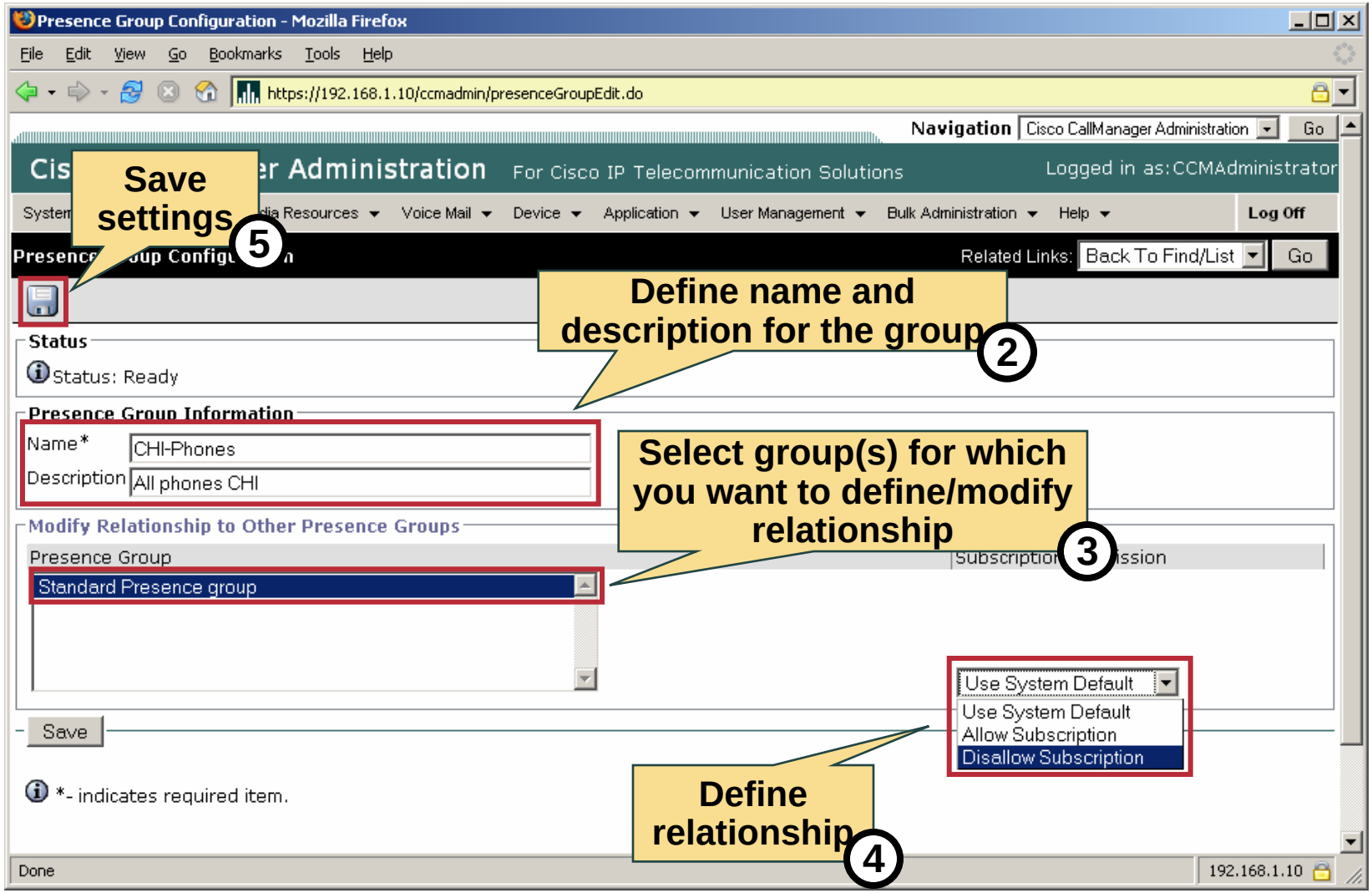
Configure Presence Authorization



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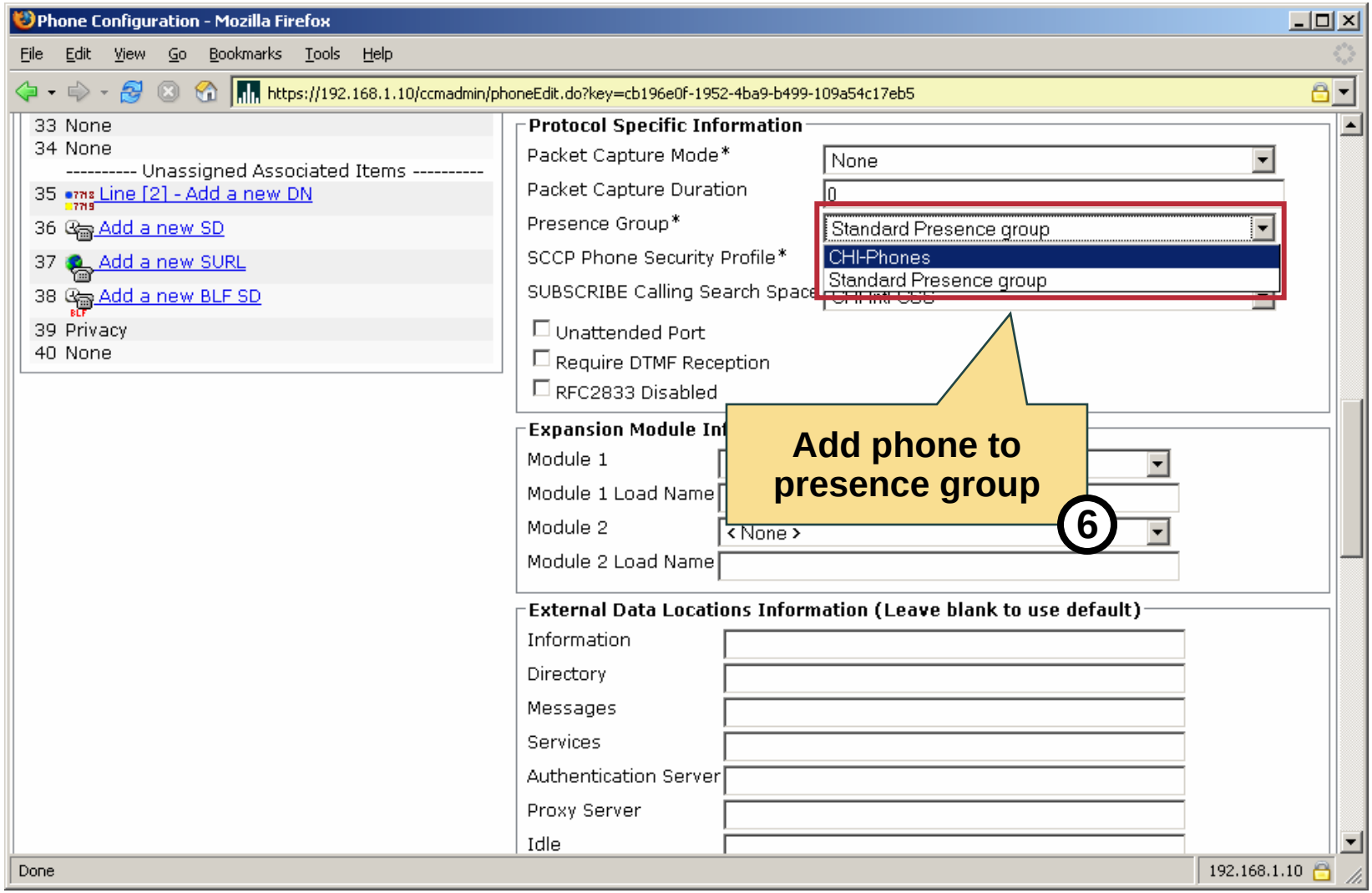
Configure Presence Authorization



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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.

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Cisco Unified CallManager 5.0 Features and Applications

Configuring Telephony Features

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Objectives

- **Configuring Call Park**
- **Configuring Call Pickup**
- **Configuring Conferencing**

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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.

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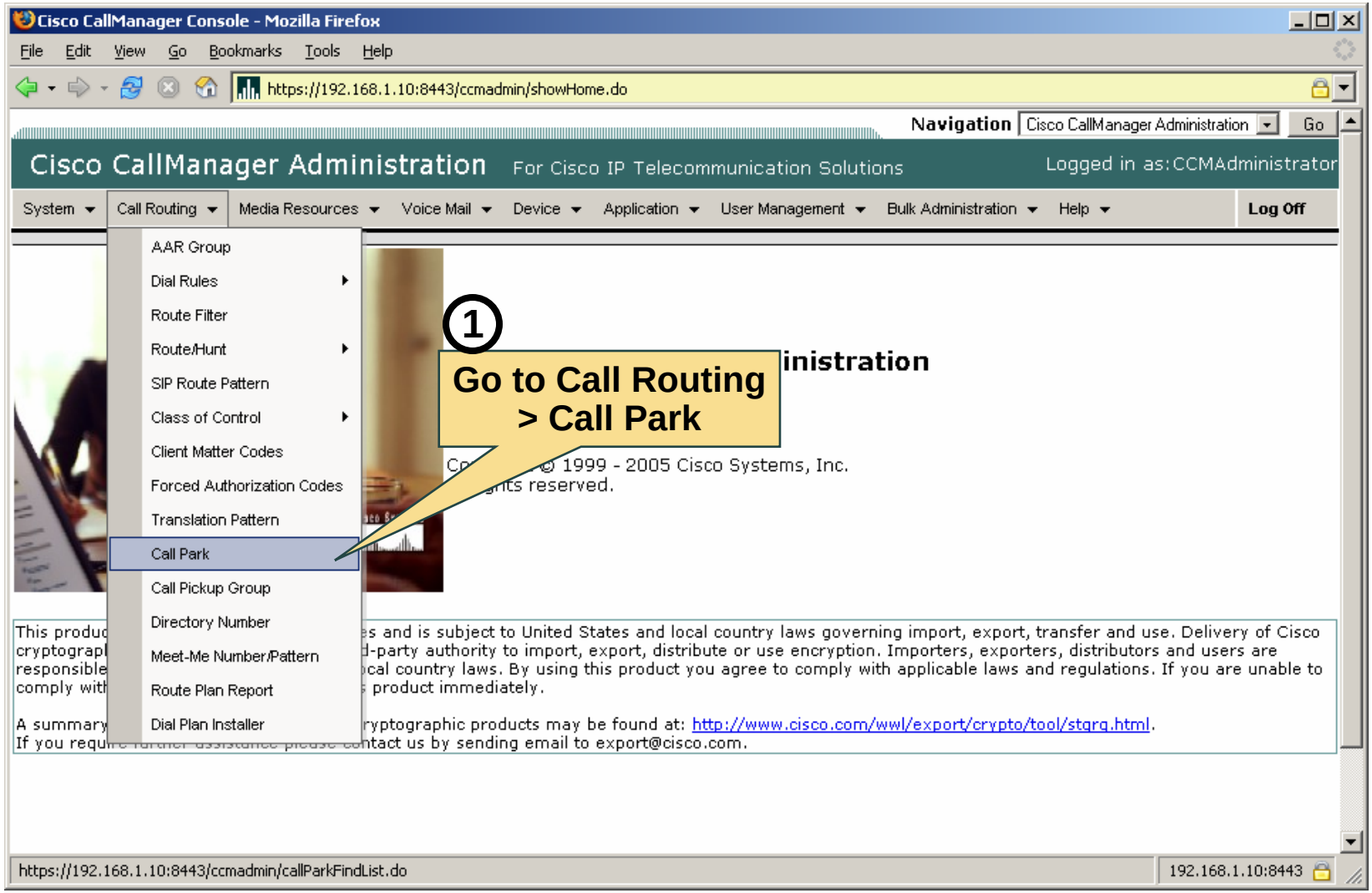
Configuring Call Park



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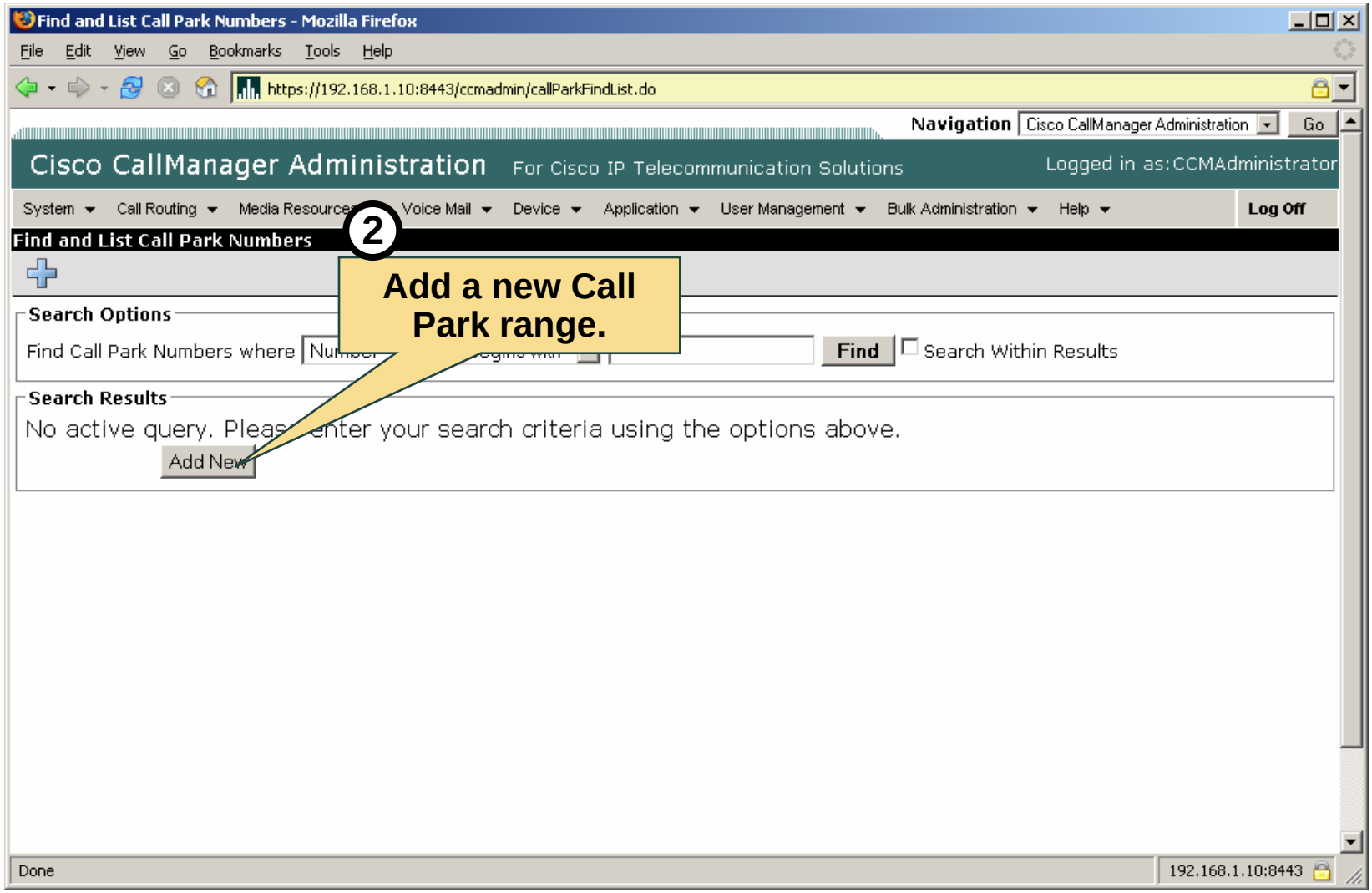
Configuring Call Park



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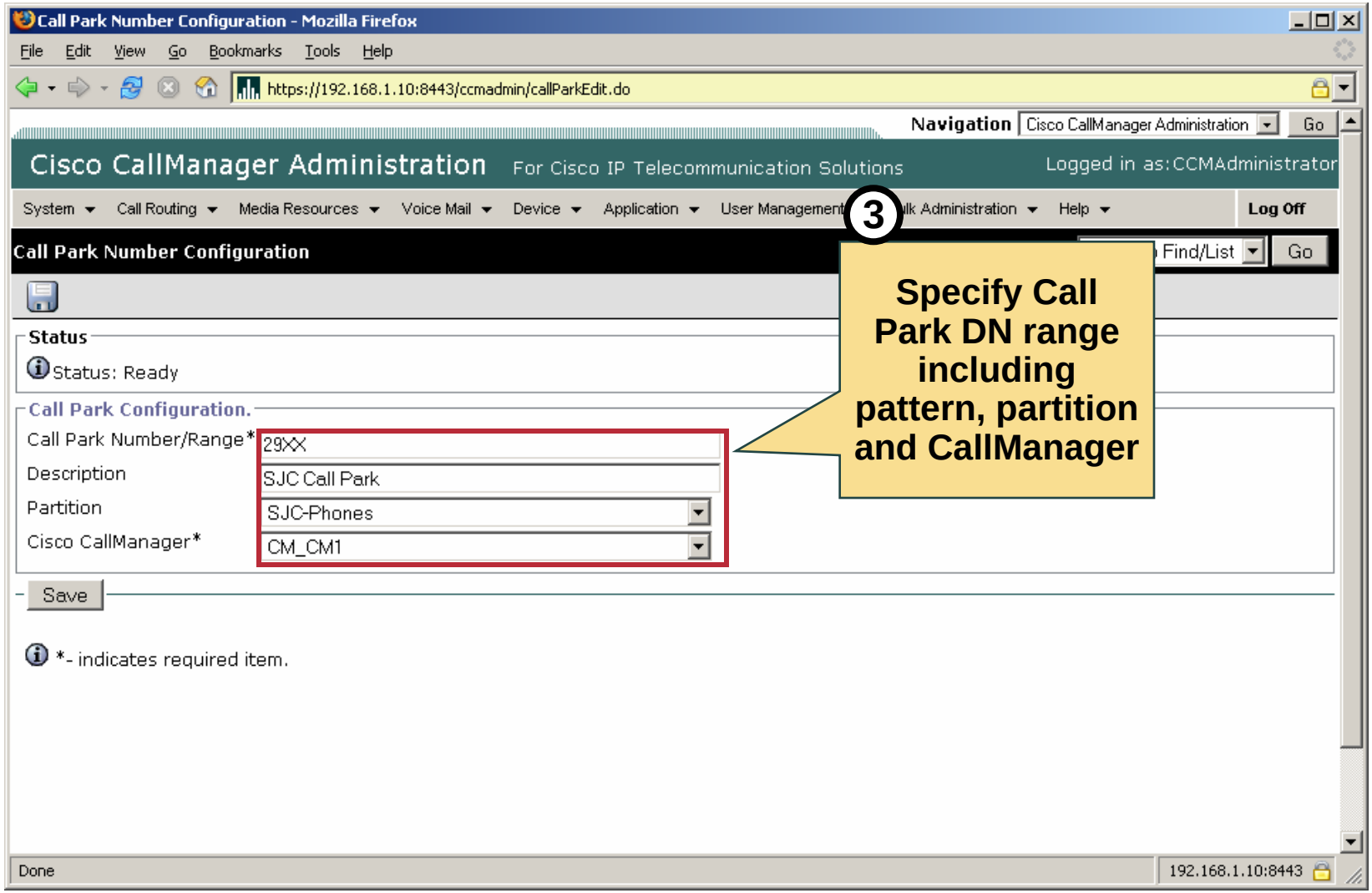
Configuring Call Park



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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.

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Configuring Call Pickup

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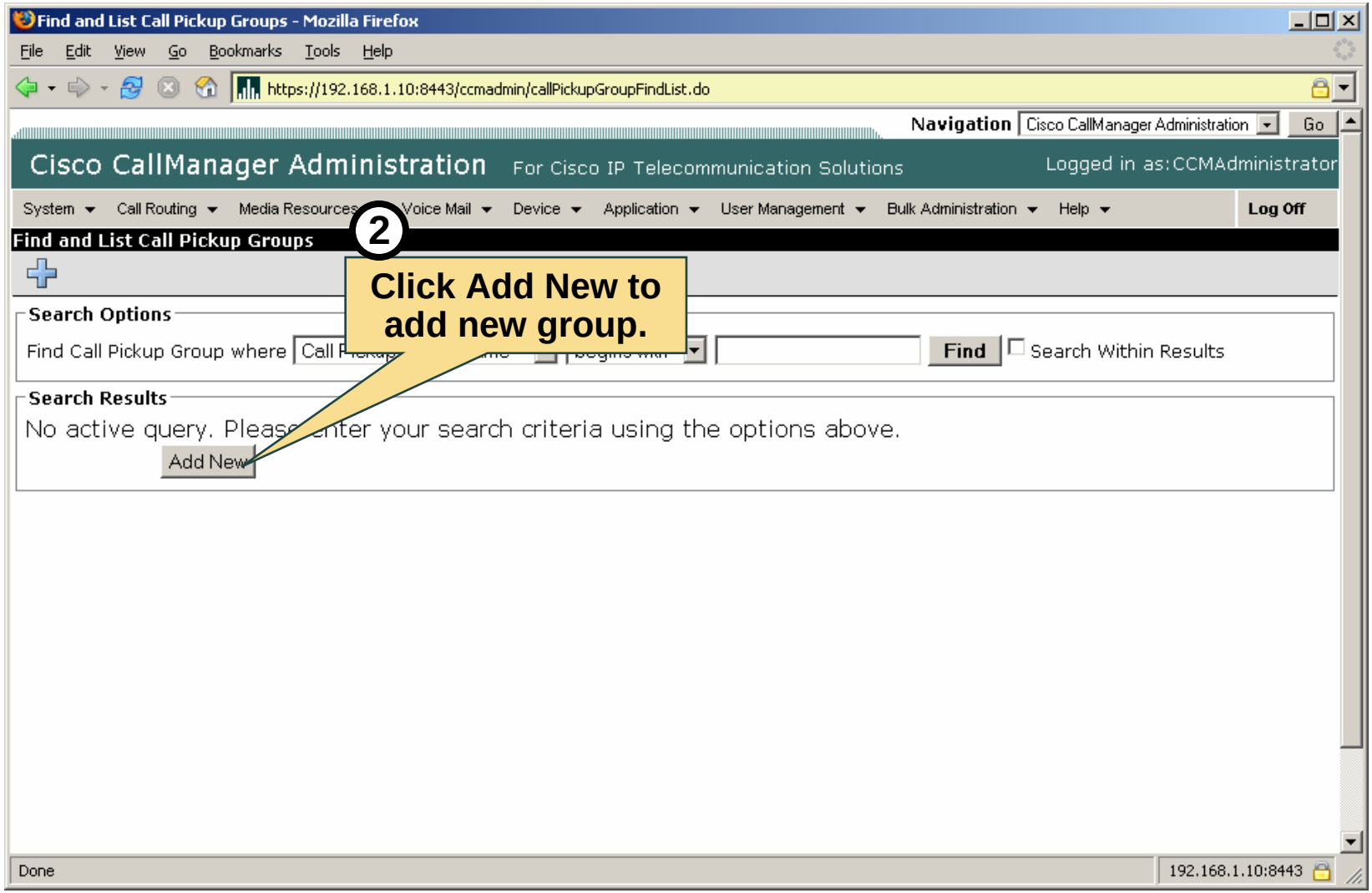
Configuring Call Pickup



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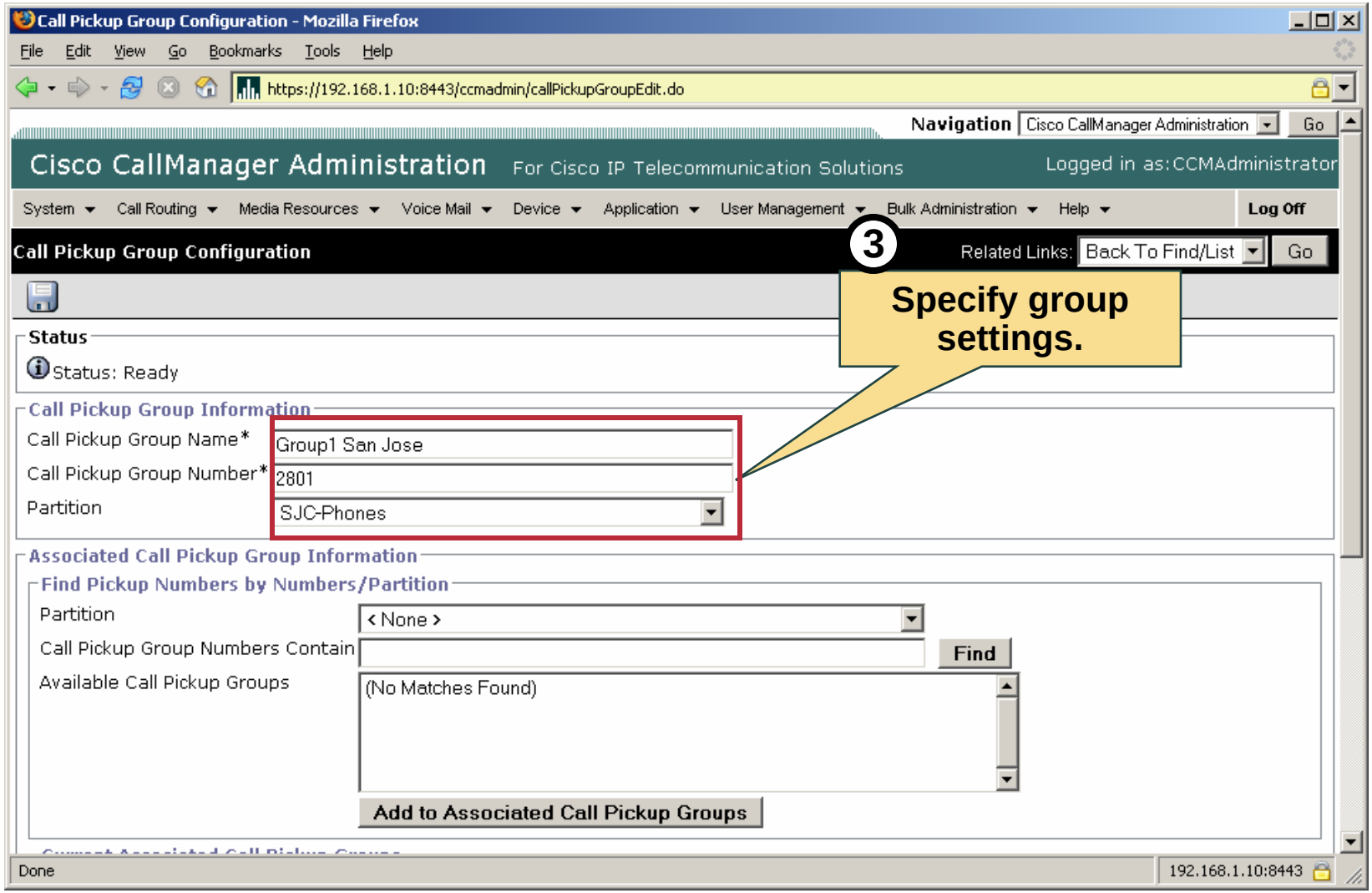
Configuring Call Pickup



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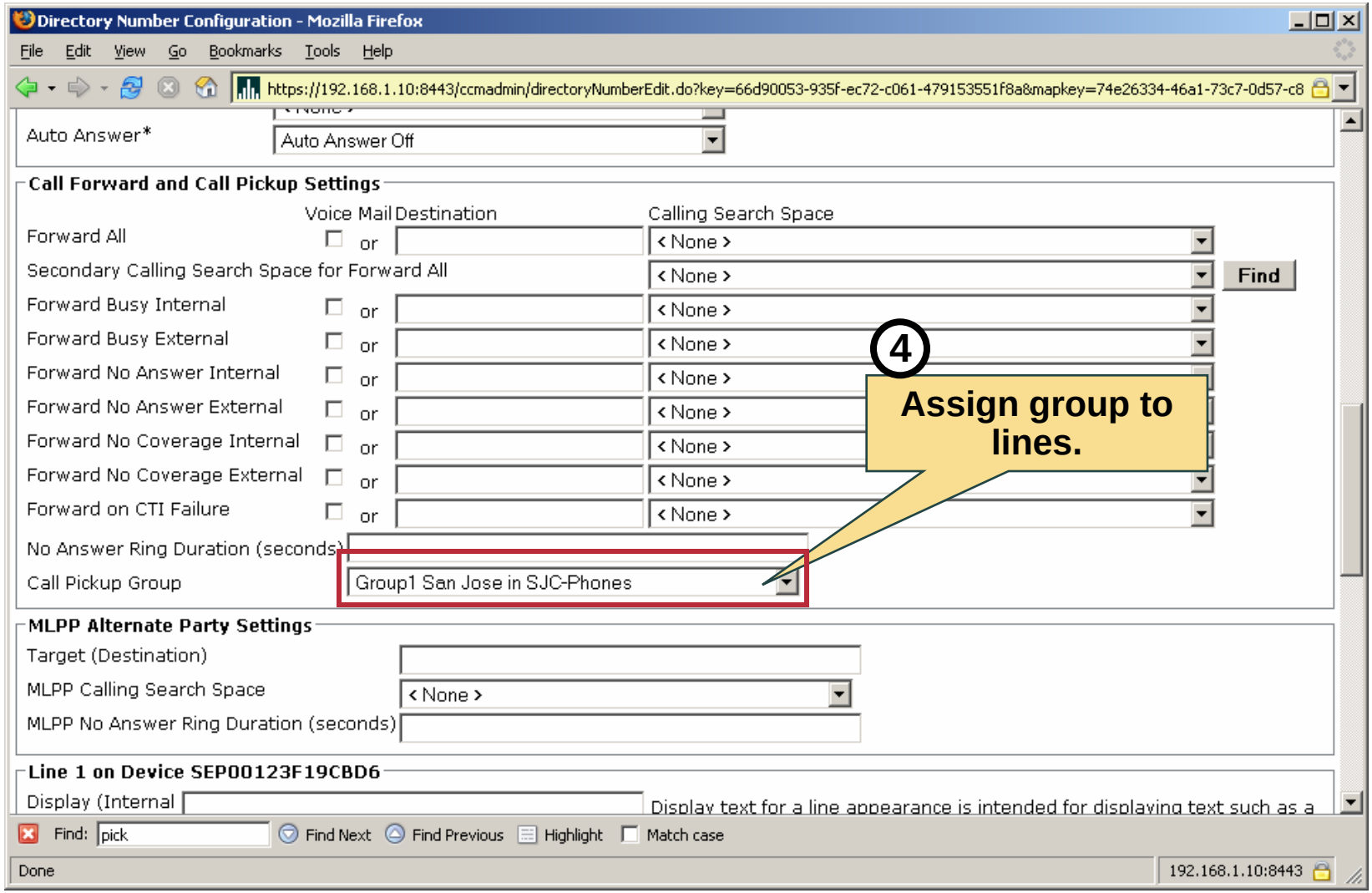
Configuring Call Pickup



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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).

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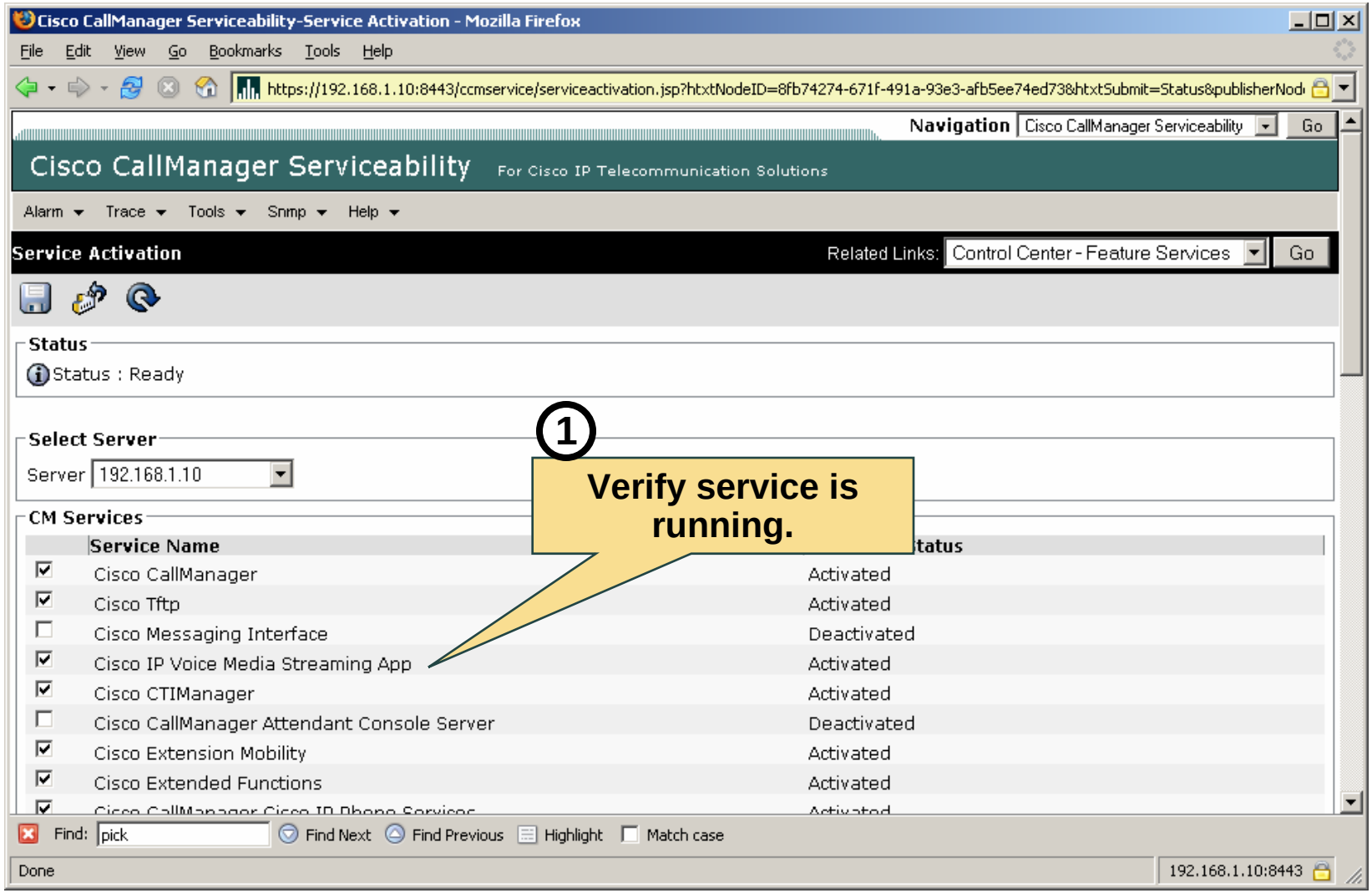
Configuring Conferencing



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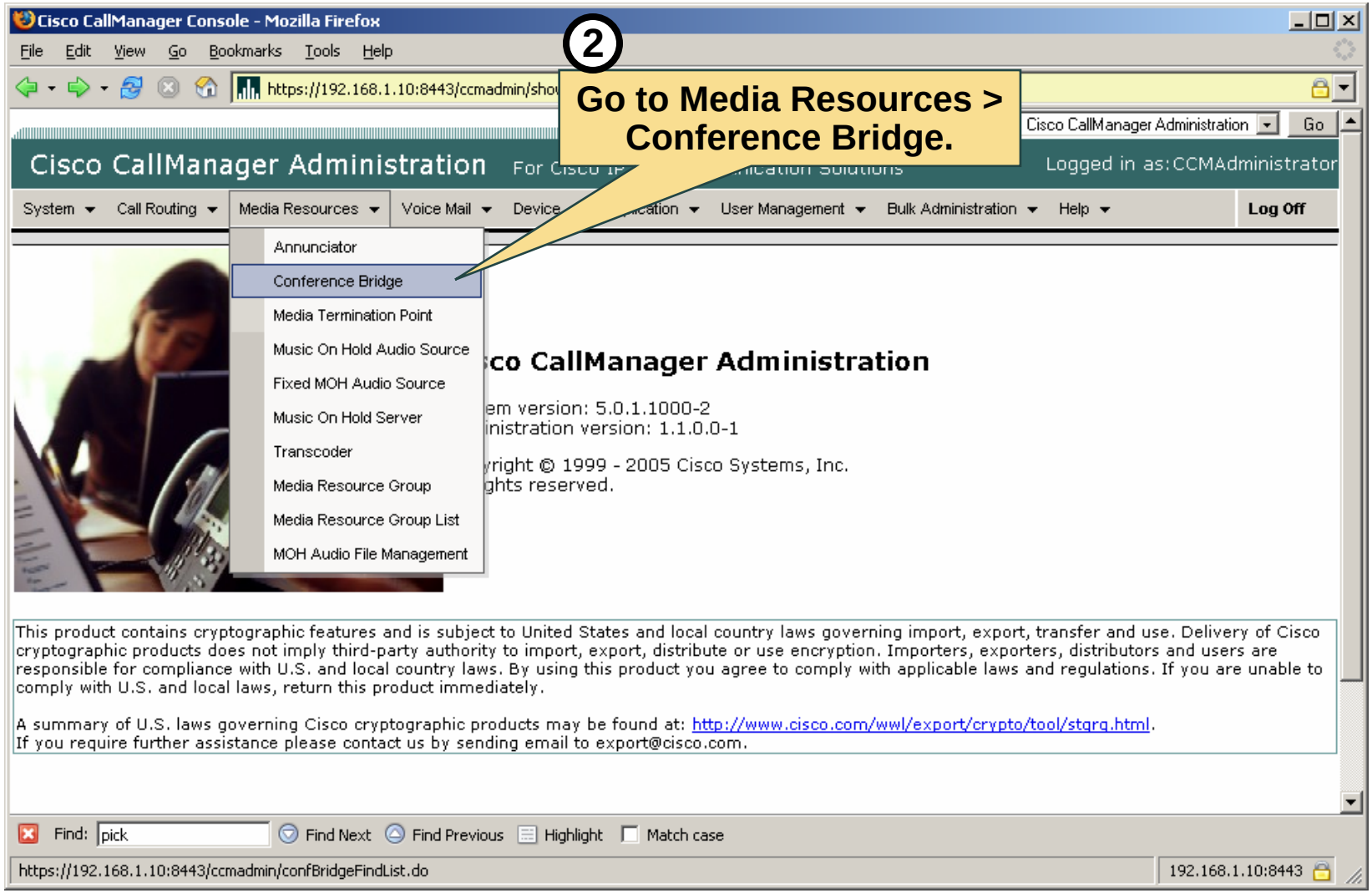
Configuring Conferencing



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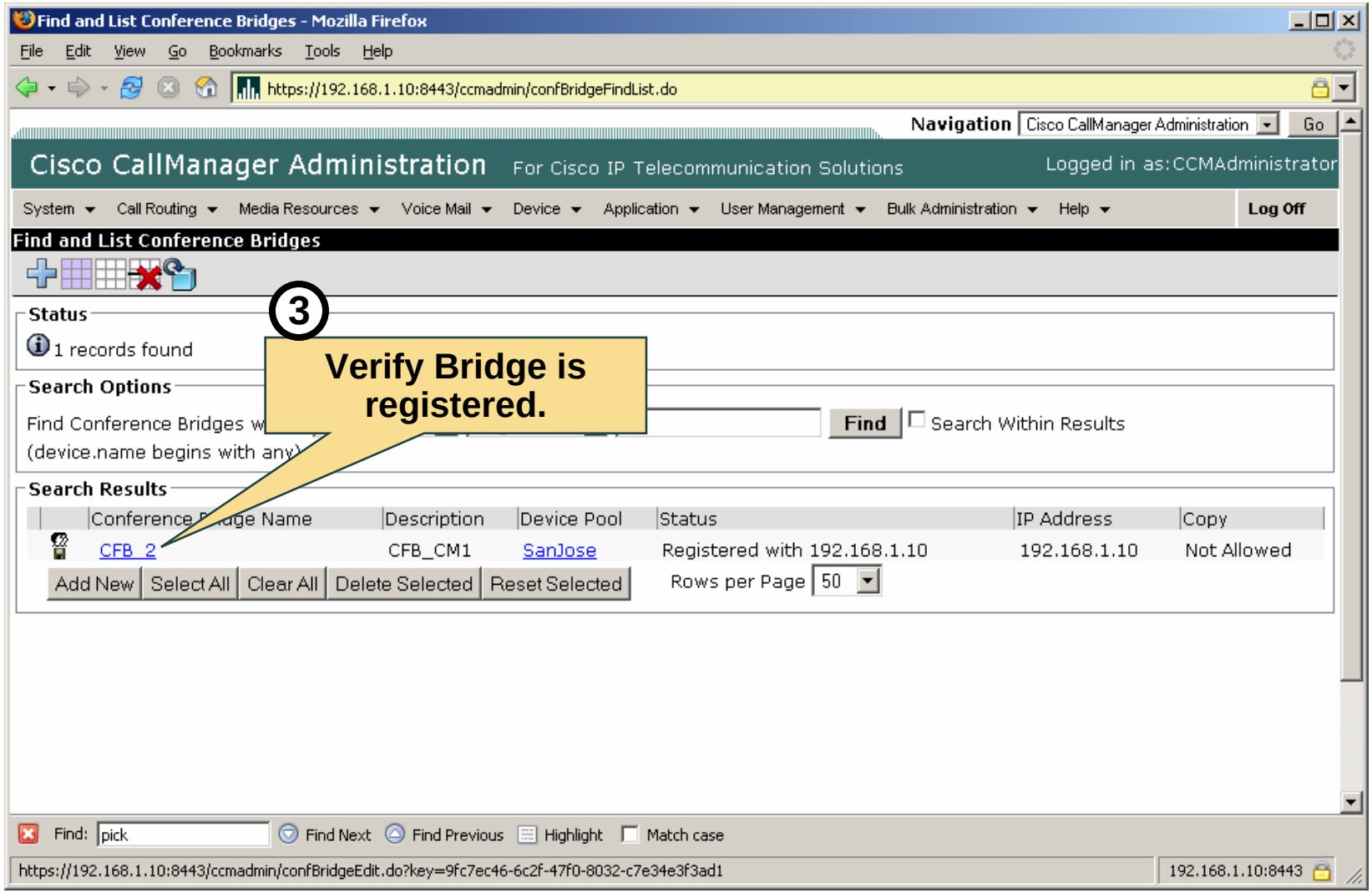
Configuring Conferencing



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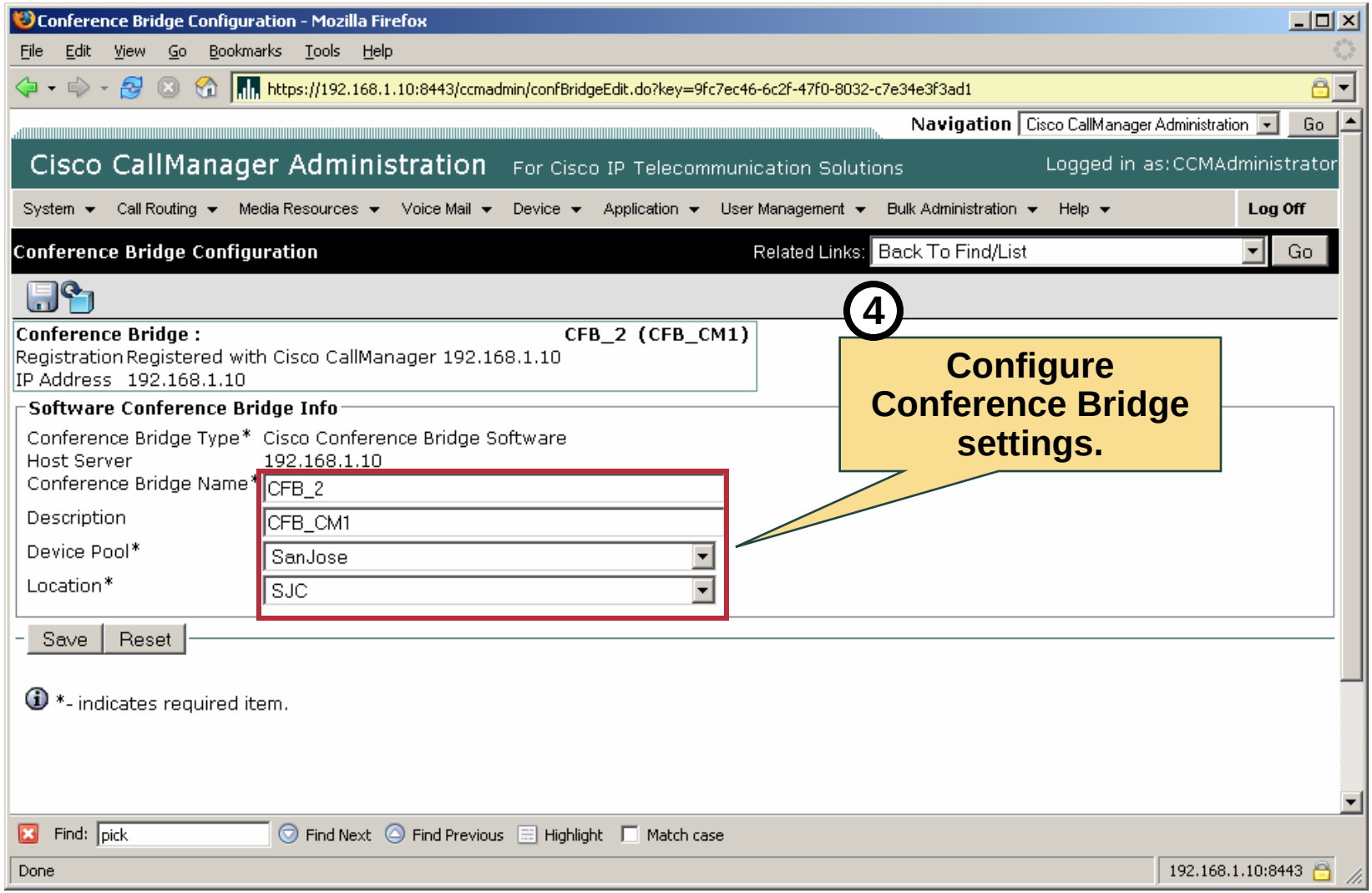
Configuring Conferencing



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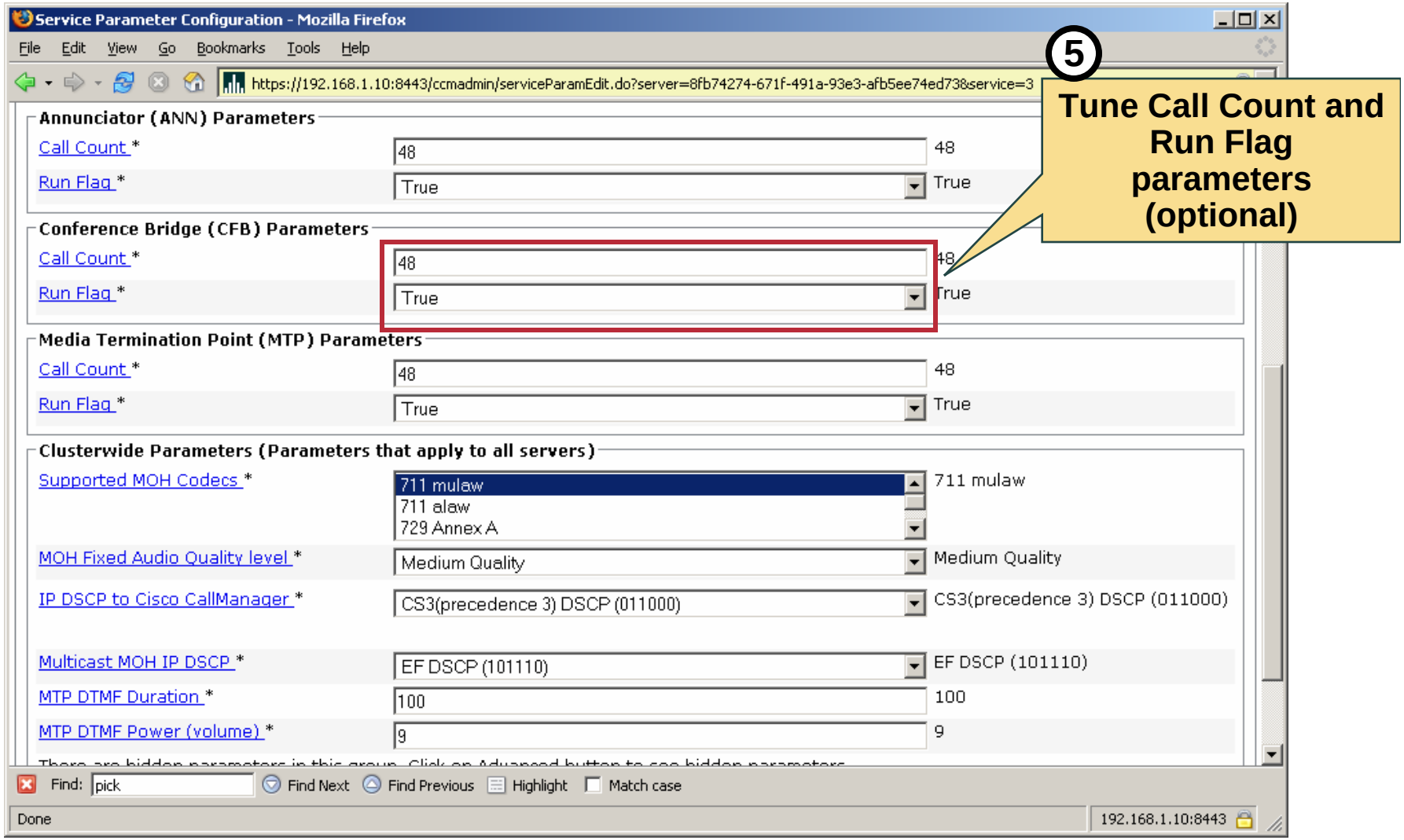
Configuring Conferencing



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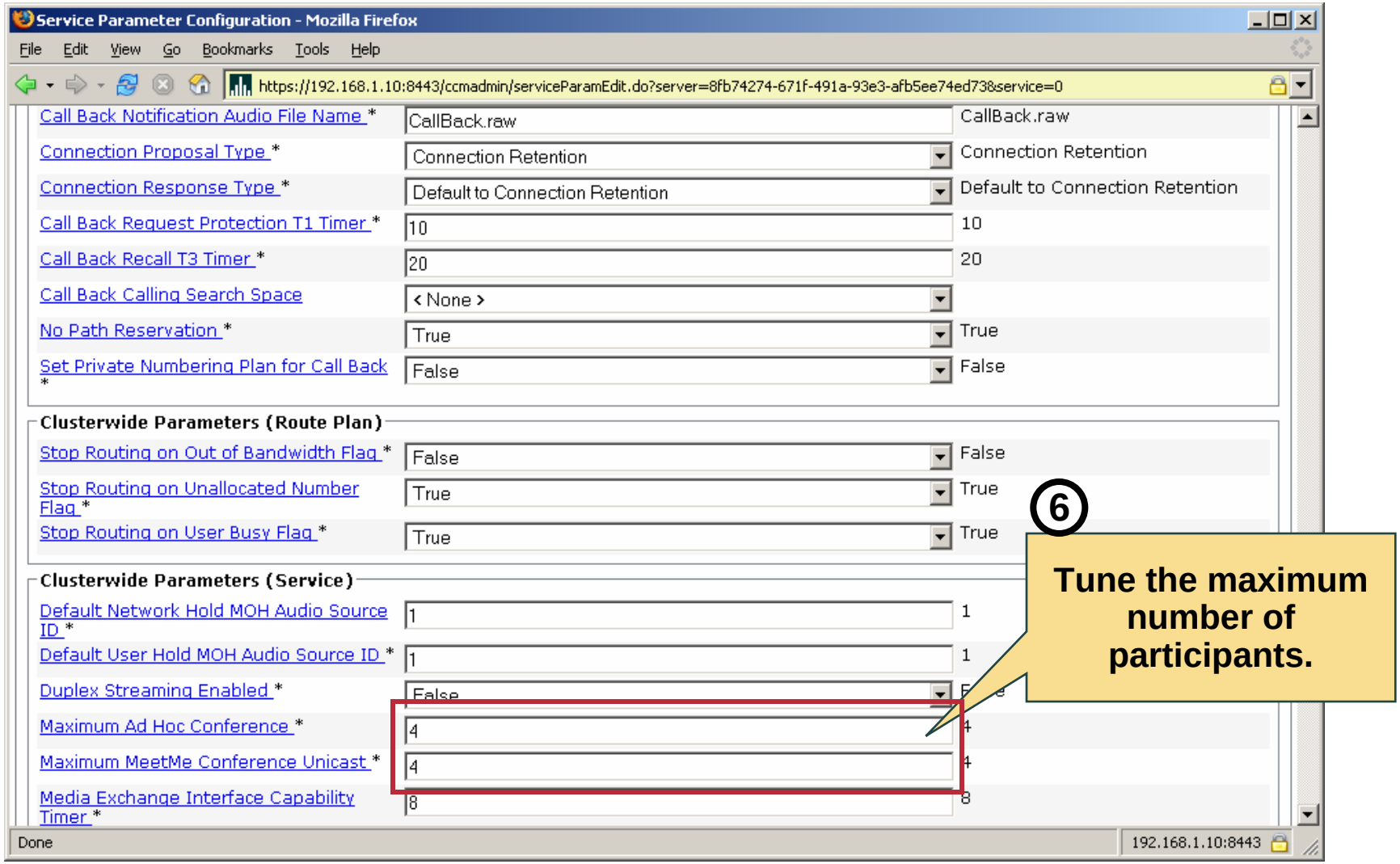
Configuring Conferencing



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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.**

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