



Cisco Collaboration Systems Release 10.5(1)

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Americas Headquarters

Cisco Systems, Inc.
170 West Tasman Drive
San Jose, CA 95134-1706
USA
<http://www.cisco.com>
Tel: 408 526-4000
800 553-NETS (6387)
Fax: 408 527-0883

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CHAPTER 1

Home

Cisco Collaboration Systems Technical Information Site



Note

From Release 10.0 onward, Cisco Unified Communications System is re-branded to Cisco Collaboration Systems to better reflect the Collaboration products included in the solution.

This information site describes the Cisco Collaboration Systems Release 10.5(1) for Collaboration (formerly known as IP telephony). Cisco Collaboration solutions integrates voice, data, and video communications together into a single network, thereby making it simple, scalable, and cost-effective solutions.

This site contains system documentation that is presented in the network lifecycle process: Prepare, Plan, Design, Implement, Operate, and Optimize (PPDIOO). PPDIOO is a Cisco methodology that defines the continuous lifecycle of services required by the customer.

Each part of the network lifecycle process has a tab at the top of the page. When you click a tab, the table of contents (TOC) on the left navigation panel changes to show only the TOC for that tab. The opening page on each tab describes what is covered in that phase. You can also use the index link at the bottom of every TOC. To learn more about how to navigate through this site, see [Using This Information System](#).

The Critical Path to Successful Deployment

The PPDIOO process is the critical path to launch and complete a successful customer deployment, from the request for information (RFI) proposal to successful training of operations personnel. The Cisco Collaboration Systems documentation is designed to be used along with the PPDIOO methodology. Each tab on the web interface contains a complete task flow for each phase of the PPDIOO process. [Audience](#) shows you what type of content you will find on each process tab.

Audience

This technical information site is designed for people who are implementing Cisco Collaboration Systems for Collaboration:

- Cisco partners
- Cisco system engineers (SEs)

- Cisco Technical Assistance Center (TAC) engineers
- Cisco customers, especially decision makers, network designers, and operations personnel

About This Release

This documentation covers a Cisco Collaboration Systems for IP telephony Release 10.5(1). If you are upgrading an existing Cisco Collaboration Systems application, begin by reading the [Collaboration System Release Notes for Cisco Collaboration Systems Release 10.5\(1\)](#) to familiarize yourself with functionality in this new release.

There are two technical information sites for Cisco Collaboration Systems Release 10.5(1):

- [Collaboration System Technical Information Site](#)
- [Contact Center System Technical Information Site](#)

Using This Information System

This information system is designed to give you an easily navigable framework for accessing all documentation for your system, solution, or product. The following topics describe using the information system:

- [About the Technical Information Site Window](#)
- [Types of Topics](#)
- [Site Index](#)
- [Graphics with Hotspots and Popup Text \(Image Maps\)](#)
- [Where Information Is Located](#)
- [About the Secondary Browser Window](#)
- [Tips on Navigating the Information Site](#)

**Note**

Make sure your browser does not block popup windows for this site. If a popup link fails to open, check your browser settings. Alternatively, press **Ctrl** when you click the link to override your browser's settings.

About the Technical Information Site Window

The window is laid out so that you can easily navigate between topics, drill down to get detailed information, and directly access product and platform documentation, without ever losing your place or having to cope with a complex hierarchy of windows.

[Figure 1-1](#) shows an example of a window for a Cisco Collaboration Systems solution. [Table 1-1](#) describes the numbered window elements.

Figure 1-1 Example of Technical Information Site Window

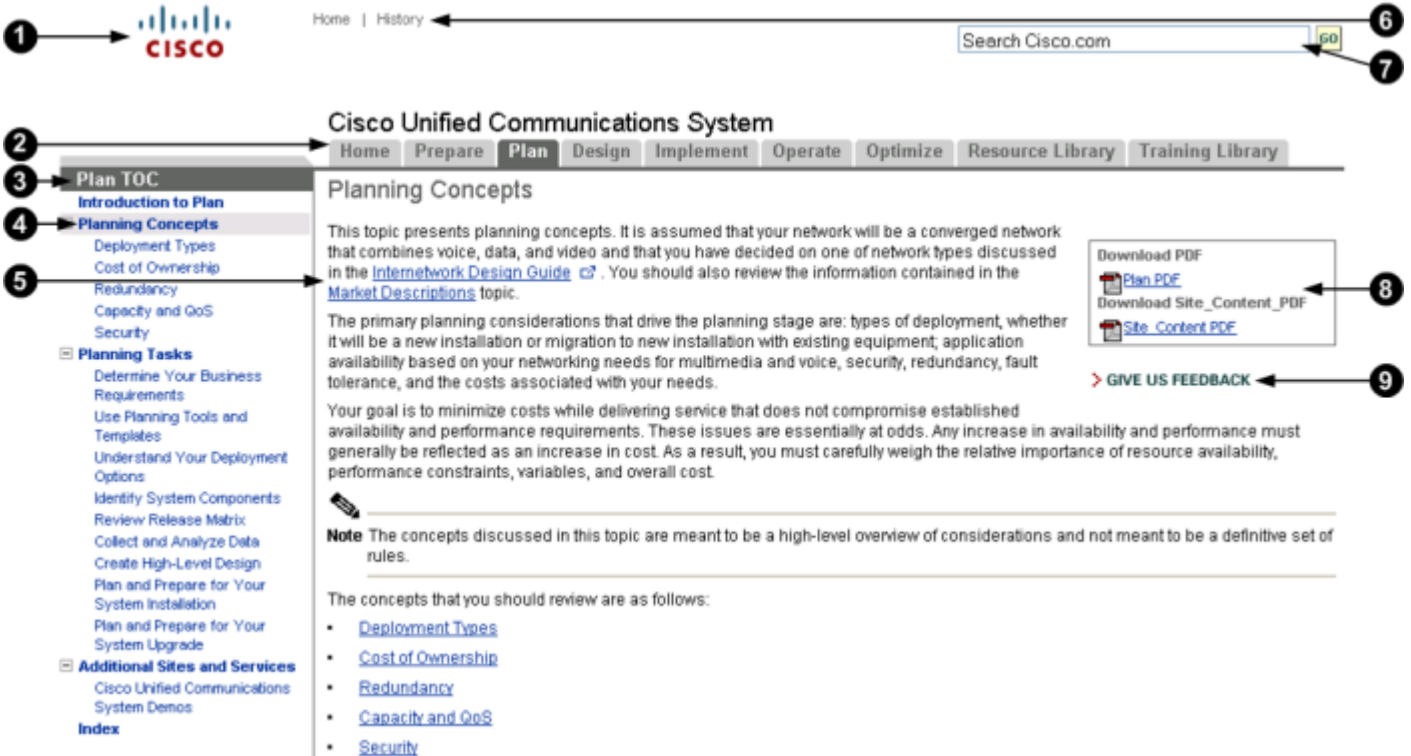



Table 1-1 Key to Window Illustration

1	Cisco logo. Click to go to the Cisco.com home page. Click the browser Back button to return to the information site window.	6	Access-from-anywhere links to Home and History.
2	Tabs for global navigation between processes or other major categories. Click a tab to go to the home page for that tab. The table of contents (TOC) changes, showing topics specific to that tab. The first content pane on a tab shows an overview of what is on the tab and the tasks and concepts covered.	7	Use the Search box to search all of Cisco.com, not specifically this information site. The search list appears in a new window so that you do not lose your place within the information site.
3	TOC for navigation within a tab. The TOC changes when you click a different tab, or when you click a link that goes to a topic on a different tab. Click the Index link at the bottom of the TOC if you are not sure where to find a topic.	8	Download an Adobe Acrobat PDF of the content of the current tab or the content of the entire site.

Table 1-1 Key to Window Illustration (continued)

4	Main heading in a TOC, such as "Planning Concepts." A blue heading links to a topic in the content pane. A black heading is unlinked and simply a title for linked subtopics below. A highlighted heading in the TOC indicates the current topic displayed in the content pane.	9	GIVE US FEEDBACK: Click to go the Feedback form at the bottom of the page to provide page-specific feedback.
5	Content pane, where the information resides. Note two kinds of links in the content pane: <ul style="list-style-type: none"> • A link to another topic in the content pane looks like an ordinary link. Clicking the link switches the contents of the pane. • A link to a secondary topic is appended with a popup icon . Clicking the link opens a new browser window, offset from the current window. If the other window is already open, the topic replaces the current contents. 		

Types of Topics

When you see a reference to a topic, you can tell what type of topic it is by its name:

- “Doing” topics, such as “Performing a System Upgrade,” are *task topics*, and provide instructions for doing something.
- “Overview” or “About” topics are *concept topics* to help you understand and plan your deployment and carry out tasks knowledgeably.

Some tabs may group topics under headings such as “Planning Concepts” and “Planning Tasks.”

Site Index

Click the Index link at the bottom of a TOC to view a hyperlinked index to all the topics in the information site. Use this index if you are not sure where to find a topic.

If a topic appears only once in the site, the index displays the entry as a clickable link. If a topic appears more than once, the entry is followed by clickable numbers linked to successive occurrences, similar to a series of page numbers in a printed index. For example,

Visio diagram, [Figure 1-1](#)




Click [Figure 1](#) to go to the relevant figure. You can use the browser Back button to return to your place in the index.

Graphics with Hotspots and Popup Text (Image Maps)



Some graphics may be image maps. An image map may have hotspots that you can run your pointer over to view a popup description or that you can click to open a linked topic in a secondary window.

Where Information Is Located

Cisco systems and solutions encompass a range of products and technologies, and their documentation encompasses information that may reside in several locations:

- Overviews and high-level process and procedure information specific to your solution or system are included directly in the information site.
- Product and technology overviews, detailed requirements, task details, and other more generic topics are located outside the site. These topics have the appearance of standard Cisco documentation with which you may already be familiar. Links to these topics appear with a popup icon  appended, for example, [Performing Your System Installation](#) . Clicking the link opens the topic in a new, secondary browser window offset from the current window, rather than replacing the current topic in the content pane. You can click the link to view the information when you need it, and then return to your place in the information site.
- Links with this symbol () are available only to people with a Cisco login, such as Cisco partners or registered Cisco.com users with a [Cisco service contract](#). After clicking the link, log in when prompted. A secondary browser window opens. Keep the secondary window open in order to open other links without having to log in again.
- Links with **[Internal]** are available only to Cisco employees.

About the Secondary Browser Window

When a topic like [Performing Your System Installation](#)  opens in a new, secondary browser window, that window stays open until you close it. (Click the **Close** button or choose **File > Close**.) If the window is open when you click another  link, the new topic replaces the current one. You can use the browser Back button if you want to retrace your steps in the secondary window.

Tips on Navigating the Information Site

- Use tabs to navigate between major process areas.
- Use the TOC at the left of the site window to navigate to major topics on a tab.
- In a secondary popup window:
 - When you are done with the window, click the **Close** button to close it. (It does not close automatically.)
 - You can go back to a previous topic by right-clicking and clicking **Back**.
 - You can view normal browser toolbars, the address bar, and any other browser items that you do not see by using commands on the View menu.
- Use the Index (click the link at the bottom of any TOC) if you are not sure where to find a topic.

Cisco Documentation

For information on obtaining documentation, support, providing documentation feedback, security guidelines, and also recommended aliases and general Cisco documents, see the monthly *What's New in Cisco Product Documentation*, which also lists all new and revised Cisco technical documentation, at:

<http://www.cisco.com/en/US/docs/general/whatsnew/whatsnew.html>

Subscribe to the What’s New in Cisco Product Documentation as a Really Simple Syndication (RSS) feed and set content to be delivered directly to your desktop using a reader application. The RSS feeds are a free service and Cisco currently supports RSS Version 2.0.

Site Content Map

The Site Content Map shows the content on each process tab in this technical information site.

Prepare	Plan	Design	Implement	Operate	Optimize
Cisco Collaboration Systems Description Market Descriptions Product Categories IP Telephony Deployment Models System Features in This Release	Deployment Types	Introduction to Design	Order Equipment	Introduction to Operating the System	Optimizing Your System
	Determine Your Business Requirements	Using Design Tools and Templates	Introduction to Implementation	Backing Up and Restoring Components	
	Use Planning Tools and Templates	Identify the Components That You Need	Install and Configure System Components	Using Network Monitoring Tools	
	Understand Your Deployment Options	Review System Caveats	Introduction to Troubleshooting System	Troubleshooting Daily Operations	
	Create High-Level Design	Develop Traffic Engineering Specifications	Troubleshooting Methodology	Resource Library	Training Library
	Preparing for Your System Installation	Define Security Policies	Conduct User Acceptance Test	System Release Documentation	General Training
	Preparing for Your System Upgrade	Design for High Availability		Solution Reference Network Design	Training Available to Partners
				System Release Notes	Training Available to Cisco Employees
				Network Topology Diagrams	
				Configuration Command Files	



CHAPTER 2

Prepare

Introduction to Prepare

In the Prepare phase, you evaluate Cisco technologies that address your business needs. Gather information about your business and technical environment that will feed into the high-level design. Then, create a business case for the IP telephony system that provides the best return on your investment.



Tip

You can navigate to any topic on this tab by using the tab navigation pane at the left of the content pane. This navigation pane contains the table of contents (TOC) for the active tab.

Before You Begin

Understand the features and functions of IP telephony applications. Start with the high-level information in the [Cisco Collaboration Systems Description](#) and [Collaboration Overview](#), and then proceed to the more detailed and Collaboration release-specific information in the [System Release Notes](#).

When You Are Done

You have defined and created the following:

- Your business and system requirements
- A basic list of components and applications that match the requirements

Major Concepts and Tasks in This Process

- [Cisco Collaboration Systems Description](#)
- [Collaboration Overview](#)

Cisco Collaboration Systems Description

Cisco Collaboration Systems offer many features and solutions that allow people to collaborate effectively. It offers a coherent experience that connects people with people in familiar, intuitive, natural, simple ways using any media or device, at any time, while integrated with business processes. This integrated collaboration architecture with a converged voice, video and data network includes many products from Cisco's Collaboration portfolio.

Enterprise, mid-market or small and medium businesses can implement various system deployment models such as single site, multiple sites or cloud. Cisco Collaboration Systems Release 10.5(1) is backwards compatible with Collaboration Systems Release 10.0(1).

For highlights of the Cisco Collaboration Systems Release 10.5(1) see:
http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/uc_system/V10-5-1/System-Description/VTGS_BK_C38BE930_00_cisco-collaboration-system-description-release/VTGS_BK_C38BE930_00_cisco-collaboration-system-description-release_chapter_00.html

Collaboration Overview

The Cisco IP telephony system includes a wide array of hardware and software components, such as call processing products, communications endpoints (Unified IP phones and video devices), and special applications, all deployed over a converged network infrastructure. The network infrastructure for Cisco IP telephony includes PSTN gateways, analog phone support, and digital signal processor (DSP) farms.

The following topics are described:

- [Market Descriptions](#)
- [Product Categories](#)
- [IP Telephony Deployment Models](#)
- [System Features in This Release](#)

Market Descriptions

Cisco provides Unified Communications solutions for small businesses through large enterprise networks. For more information on applying Unified Communications solutions to different size markets, see the following:

- [Cisco Unified Communications Solutions for Enterprise](#)
- [Cisco Unified Communications Solutions for Small Business](#)

Product Categories

IP telephony system components are grouped in the following categories:

- Call control components, such as Cisco Unified Communications Manager, Cisco Unified Communications Manager Express, Cisco Unified Survivable Remote Site Telephony (Unified SRST), Cisco TelePresence Video Communication Server, Cisco Business Edition 7000 and Cisco Business Edition 6000.
- Server applications, such as Cisco Unified Communications Manager IM and Presence Service, Cisco Emergency Responder, Cisco Paging Server, Cisco Unified Attendant Console Standard, and Cisco Unified Attendant Console Advanced.
- Conferencing components, such as Cisco TelePresence Server, Cisco TelePresence MCU, Cisco TelePresence Conductor, Cisco WebEx Meetings Server, and Cisco TelePresence Management Suite.
- Voicemail and unified messaging components, such as Cisco Unity Connection.
- Devices (endpoints) and clients, such as Cisco IP Phone 7821, 7841, 7861, Cisco Unified IP Phones 6900, 7800, 7900, 8900, and 9900 Series, Cisco TelePresence SpeakerTrack 60, Cisco Desktop Collaboration Experience DX650, Cisco TelePresence SX10, and Cisco TelePresence MX200 G2.

- Client applications such as Cisco Jabber for Android, Cisco Jabber for Mac, Cisco Jabber for iOS, and Cisco Jabber for Windows.
- Network Management devices, such as Cisco Prime Collaboration Provisioning, Cisco Prime Collaboration Provisioning, and Cisco Prime Collaboration Deployment.
- Communications infrastructure devices, such as Cisco IOS Extended Maintenance Deployment, Cisco 4451-X Integrated Services Router (ISR), and Cisco SRE 910 Service Module.

IP Telephony Deployment Models

Cisco Collaboration Systems for IP telephony supports the deployment models in [Table 2-1](#).

Table 2-1 Deployment Models

Deployment Model	Description
Single-site Model	This model is designed for autonomous offices in which most or all employees are IPC users. This model can support up to 30,000 users.
Multisite Centralized Call Processing Model	This model is designed for distributed operations with a large central or headquarters site and multiple remote or branch sites. This model can support up to a total of 30,000 phones distributed among up to a maximum of 1000 sites. Based upon the bandwidth available, each site can support any number of users up to the overall total of 30,000 phones.
Multisite Distributed Call Processing Model	This model is designed for organizations with large user populations or large numbers of geographically distributed sites resulting in the need for more than a single call processing entity. This model is suited for deployments that require multiple Cisco Unified Communications Manager clusters or Cisco Unified Communications Manager Express platforms. Each call processing entity in this model is configured as a Single-Site Model or Multisite Centralized Call Processing Model and each has a common dial plan and feature set.
Clustering Over IP WAN Call Processing Model	This model is designed for organizations with large user populations across multiple sites that are connected by an IP WAN with the QoS features enabled. It supports the Local Failover Deployment Model and the Remote Failover Deployment Model.

See also [Deployment Methodology](#) in the Cisco Collaboration Systems Description.

System Features in This Release

Cisco Collaboration Systems integrate telephony, conferencing, messaging, and security products for IP customers who have a variety of deployment models. For detailed Cisco IP telephony feature information, see the [System Release Notes: Cisco Collaboration Systems, Release 10.5.1](#).

Additional Product Information

[Cisco Unified Enterprise Solution](#)



CHAPTER 3

Plan

Introduction to Plan

In the Plan phase, you assess your readiness to support a proposed solution. Planning continues the needs analysis begun in the Prepare phase, with the goal of producing a high-level project plan and the initial site survey.



Tip

You can navigate to any topic on this tab by using the tab navigation pane at the left of the content pane. This navigation pane contains the table of contents (TOC) for the active tab.

Before You Begin

Understand the features and functions of IP telephony applications. Start with the [Planning Concepts](#) and the [System Release Notes](#). Then review the business requirements, deployment models, and sites to understand the options that are available for your specific environment.

When You Are Done

You have defined and created the following:

- A comprehensive list of components and applications that match the requirements
- A project plan based on those requirements including a proposed, high-level design

Major Concepts and Tasks in This Process

- [Planning Concepts](#)
- [Planning Tasks](#)

Planning Concepts

This topic presents planning concepts. It is assumed that your network will be a converged network that combines voice, data, and video and that you have decided on one of network types discussed in the [Internetwork Design Guide](#). You should also review the information contained in the [Market Descriptions](#) topic.

The primary planning considerations that drive the planning stage are: types of deployment, whether it will be a new installation or migration to new installation with existing equipment; application availability based on your networking needs for multimedia and voice, security, redundancy, fault tolerance, and the costs associated with your needs.

Your goal is to minimize costs while delivering service that does not compromise established availability and performance requirements. These issues are essentially at odds. Any increase in availability and performance must generally be reflected as an increase in cost. As a result, you must carefully weigh the relative importance of resource availability, performance constraints, variables, and overall cost.

**Note**

The concepts discussed in this topic are meant to be a high-level overview of considerations and not meant to be a definitive set of rules.

The concepts that you should review are as follows:

- [Deployment Types](#)
- [Cost of Ownership](#)
- [Redundancy](#)
- [Capacity and QoS](#)
- [Security](#)

Deployment Types

The deployment types to consider are as follows:

- New Installation
 - Greenfield—Completely new installation of the Cisco Collaboration system, using no existing equipment.
 - Legacy—New installation of the Cisco Collaboration system combined with existing legacy equipment, such as TDM PBXs and third-party adjuncts, which may require long-term co-existence and integration or eventual migration to the new installation.
 - Brownfield—Existing Cisco Collaboration system, which requires an upgrade and migration from a previous system release to the current system release.
- Single-Stage Upgrade
 - Using existing hardware—All components in the network start at the base release set and all components can be upgraded to the target release set within a single maintenance window.
 - Using new hardware (flash-cut or shrink-and-grow)—A parallel network should be built using new hardware and pre-staged with configuration to support the existing production network.
- Multistage System Upgrade
 - Using existing hardware (hybrid system)—The components in individual sites can be upgraded from the base release set to the target release set in stages, during separate maintenance windows.
- Multisite Migration with Independent Site Upgrade
 - Using an hybrid network with interworking release sets—Components are upgraded on a site-by-site basis during separate maintenance windows. At the completion of each maintenance window, a hybrid network exists within the multiple sites that have components operating on the base release set; or components that are operating on the target release set; or components that are a hybrid system.

Cost of Ownership

Information system budgets can run into millions of dollars. As large organizations increasingly rely on electronic data for managing business activities, the associated costs of computing resources continue to rise. With this in mind, your basic network plan should include the following:

- Environmental consideration—Include the location of hosts, servers, terminals, and other end nodes; the projected traffic for the environment; and the projected costs for delivering different service levels.
- Performance constraints—Consider network reliability, traffic throughput, and host and client computer speeds. For example, network interface cards and hard drive access speeds.
- Internetworking variables—Include the network topology, line capacities, packet flow assignments, redundancy and fault tolerance factors, backward compatibility (co-existence and interoperability), and security.

Redundancy

Redundancy is critical considering the number of vital business applications running on the network. If you have a distributed network with several access layers to remote offices, and you have a failure from the distribution layer to the core without redundancy, you have loss of network service for a large number of people. If you have redundancy in the distribution layer and the core, you can potentially lose one or more circuits without disturbing service to any particular group of users. Depending on the application, you may also need some redundancy from the access layer to the distribution layer.

Because of redundancy, if you drop a link at any one point in the network, every remote group or user still has a path to get back to the core. Even if you cut off the connection from one of the distribution switches back to the core, you still have access to the core for every user.

For more information on redundancy planning, see the [Redundancy and Load Sharing Design Guide](#).

Capacity and QoS

Capacity and QoS are major considerations in a converged network and effect one another. QoS is needed to prevent applications from using more than a fair share of bandwidth and degrading the performance of other applications. At the WAN interface, QoS is needed to allocate expensive wide area capacity among applications.

Bandwidth and QoS requirements are easy to figure in a multilayered design because the traffic flow is fairly predictable. You can also have end-to-end QoS in a multilayered design. End-to-end QoS is critical when you have real-time applications, such as a voice conversation or video presentation, and you have non-real time applications that can interfere with the real-time applications. For example, if the real-time and non-real time applications arrive at the same layer at the same time, the network must pass the real-time packets first, as well as keep latency and jitter low. QoS end-to-end is the answer.

Consider Call Admission Control (CAC) as an alternative to QoS. CAC limits the amount of traffic allowed onto the network at the ingress point. Because you know that the network will be congested at various times during the day, you can disallow additional traffic by using CAC. Also consider using traffic shaping techniques using a traffic shaping devices. A combination of QoS, CAC and traffic shaping will provide optimal performance for applications on a converged network.

Managing link speed mismatches is the last element of traffic management. The mismatches, called chokepoints or bottlenecks, are a basic design issue whenever a large capacity link generates traffic destined for a low capacity link. To avoid the mismatches, carefully analyze the traffic and the device capabilities, then upgrade the interface (if needed) and apply a combination of CAC and QoS.

For more information on QoS, see the [Enterprise QoS Solution Reference Network Design Guide](#).

Security

Cisco recommends multiple layers of security technologies to prevent a single configuration error from jeopardizing the security of the network. Cisco also recommends operational processes that ensure prompt application of software patches, timely installation of new security technologies, and performance of regular security audits and assessments.

As you begin to design your network, rank the importance of your network assets and services by considering these factors:

- What keeps you in business?
- How do you make money?
- Does loss of data or privacy equal lost money?
- What about regulatory compliance?
- How do you protect your critical data?
- Where does voice fit?

Then consider the potential threats to your business, which may include:

- Toll fraud
- Eavesdropping
- Address spoofing
- Fake caller identity
- Media tampering
- Denial of service
- SPAM, SPIT (SPAM over IP telephony), and SPIM (SPAM over Instant Messaging)

In addition to the operational processes, advanced security technologies should be reviewed and considered. Security technologies can be categorized as follows:

- Network security
 - Virtual LANs (VLANs)
 - Access control lists (ACLs),
 - Stateful firewalls with protocol aware inspection
 - Virtual Private Networks (VPNs)
 - QoS
 - Dynamic Address Resolution Protocol (ARP) inspection
 - Dynamic Host Configuration Protocol (DHCP) snooping
 - Port security
 - Network intrusion prevention

- Host security
 - Cisco Security Agent
 - Third-party anti-virus software
 - Host-based firewalls
 - Hardened operating systems
- User authentication, authorization, and accounting security
 - Phone image authentication
 - Multilevel administration privileges
 - Call detail reporting

For more information about Cisco end-to-end security designs, see the Cisco SAFE guidelines at <http://www.cisco.com/go/safe>. For more details about Cisco integrated network security solutions, see the following resources:

- [Security Products and Solutions](#)
- [Secure Unified Communications](#)
- [Cisco Support Community for Security](#)

Planning Tasks

The following overview shows the high-level tasks of the planning process:

- [Determine Your Business Requirements](#)
- [Use Planning Tools and Templates](#)
- [Understand Your Deployment Options](#)
- [Identify System Components](#)
- [Review Release Matrix](#)
- [Collect and Analyze Data](#)
- [Create High-Level Design](#)

Determine Your Business Requirements

Two important factors that drive your business requirements are:

- Size of your business, see [Market Descriptions](#)
- Requirements for installation and upgrade, see:
 - [Install and Configure System Components](#)
 - [Additional Sites and Services](#)

Review [Deployment models](#) chapter in the Cisco Collaboration Systems Description Release 10.5(1) for more details.

Collecting Requirements

The following are suggested methods to use in gathering information to plan your network:

- **Assess User Requirements**—Users want applications to be available on demand in the network. The chief components of application availability are response time, throughput, and reliability. You can assess user requirements as follows:
 - Develop community profiles of what different user groups require. Although many users have roughly the same requirements of an electronic mail system, engineering groups using Windows terminals and Sun workstations in an NFS environment have different needs from PC users sharing print servers in a finance department.
 - Build a baseline for implementing an internetwork by interviewing groups, forming focus groups, or using surveys. Some groups might require access to common servers, while others might want to allow external access to specific internal computing resources. Formal surveys can be used to get a statistically valid reading of user sentiment regarding a particular service level or proposed internetworking architecture.
 - Conduct a test involving representative users in a lab environment. This is most applicable when evaluating response time requirements. As an example, you might set up working systems and have users perform normal remote host activities from the lab network. By evaluating user reactions to variations in host responsiveness, you can create benchmark thresholds for acceptable performance.
- **Identify Functionality Requirements**—After you understand your internetworking requirements, you can select the specific functionality that fits your environment, such as the level of application availability and the implementation costs for that availability. Fault tolerance and redundancy should be considered also.

Use Planning Tools and Templates

This topic includes planning tools and links to documents that provide guidelines for designing and configuring your IP telephony system. It also includes information on quoting and ordering Cisco Unified Communications products.

Solution Reference Network Design Documents

Solution Reference Network Design (SRND) documents provide guidelines, recommendations, and best practices for implementing IP telephony network solutions. The following SRNDs are recommended for designing Cisco Collaboration Systems:


- [Cisco Collaboration Systems SRND Based on Cisco Unified Communications Manager 10.x](#)
- [Cisco Unified Communications SRND Based on Cisco Unified Communications Manager 9.x](#)
- [Cisco Unified Communications SRND Based on Cisco Unified Communications Manager 8.x](#)
- [Cisco Unified Communications SRND Based on Cisco Unified Communications Manager 7.x](#)
- [Cisco Unified Communications SRND Based on Cisco Unified Communications Manager 6.x](#)
- [Cisco Unified Contact Center Express SRND Release 10.0](#)
- [Cisco Unified Contact Center Express SRND Release 9.0](#)
- [Enterprise QoS System Reference Network Design](#)

**Note**

Additional SRND resources are available at <http://www.cisco.com/go/srnd>.



Ordering Guides

[Ordering Guides](#)  for most Cisco Unified Communications products are available for Cisco partners, Cisco sales staff, and Cisco service providers.

Understand Your Deployment Options

Review the [Deployment Models](#) chapter in the Cisco Collaboration System Description for a guide to site models.

Identify System Components

For a brief description of all the components that are available with Cisco Collaboration Systems Test Release 10.5(1), refer to the [System components](#) chapter in the Cisco Collaboration System Description Release 10.5(1).

See the [Install and Configure System Components](#) topics in the Implement tab for links to information that describe components that are specific to the IP telephony system.

Review Release Matrix

The Release Matrix (which includes the [Software Version Matrix](#)) lists all the components and their versions for a particular release. This is the recommended set of components and specific software versions that have been tested and verified for interoperability within a specific system release.

Collect and Analyze Data

Using available tools, system designers collect data on the network to assess network readiness.

Tasks for data collection and analysis include:

- Perform an infrastructure analysis—By obtaining floor plans and campus maps, including utilities and conduit systems, deficiencies in infrastructure can be identified.

- Perform a software gap analysis—Do a software gap analysis to address network management tools for the IP network.
- Perform initial traffic analysis—Collect data on all potential converged infrastructure traffic flows. Use station message detail recording (SMDR) and billing records to determine legacy call volumes and use network management tools to collect key statistics on your IP data network.

Create High-Level Design

Once data is collected and analyzed, record the results in the site survey and high-level design documents.

Preparing for Your System Installation

This topic provides links to documentation for you to review before you install the Cisco Collaboration System. It lists the components in the release set and provides information regarding the deployment of various components. See the following sections:

- [Software Version Matrix](#)
- [Limitations and restrictions](#)

When your installation plans are complete and you are ready to install components, go to [Performing Your System Installation](#).

**Note**

There may be more than one upgrade path available based on the software deployed in your specific environment.

When your upgrade plan is in place and you are ready to upgrade, go on to [Preparing for Your System Upgrade](#).

Preparing for Your System Upgrade

This topic discusses information to review before the actual upgrade process, such as the different IP telephony components, upgrade release versions of components involved in the upgrade, and release version compatibility. This topic contains the following sections:


- [Upgrade paths](#)
- [Software version matrix](#)

When your upgrade plan is in place and you are ready to upgrade, go to [Performing a System Upgrade](#) on the Optimize tab.

**Note**

There may be more than one upgrade path available based on the software deployed in your specific environment.

Additional Sites and Services

Steps to Success is a Cisco methodology that outlines the tasks required to complete a successful customer engagement. Registered users can visit the [Steps to Success](#)  resource site for Cisco Unified Communications process flows.

Cisco Unified Communications Services is a Cisco service offering that provides engineering expertise and best practices.

- Registered users can visit the [Cisco Unified Communications Services](#)  partner site.
- Non registered users can visit the [Cisco Unified Communications Services](#) site.

Cisco Collaboration System Demos

Tools are available to demonstrate the collaboration features of a Cisco Collaboration System:

- For Cisco partners: [dCloud: The Cisco Demo Cloud](#) 
- For Cisco sales teams: [Global Demonstrations](#) **[Internal]**



CHAPTER 4

Design

Introduction to Design

Using the project plan that was developed in the Plan phase, your team should have enough information to develop a detailed design for each site and the entire network. The network design should contain, at a minimum:

- Routing and switching component connectivity
- Software applications and configurations for routers and switches
- Power and environment

For each site, in addition to the above, your design should include telephone circuitry, equipment racks with cabinet locations, and layouts. Each site should have a design that encompasses your network call processing, conferencing, and messaging requirements. The design should be scalable for future growth.

For an overview of the deployment options, see [Deployment models](#) in the Cisco Collaboration Systems Description Release 10.5(1).



Tip

You can navigate to any topic on this tab by using the tab navigation pane at the left of the content pane. This navigation pane contains the table of contents (TOC) for the active tab.

Before You Begin

Review Solution Reference Network Design (SRND) documents and design tools. Gather requirements and data, which can include:

- Business and system requirements
- Service-level agreements
- Capacity (bandwidth) requirements
- Site survey and proposal from the project plan

When You Are Done

The main deliverable of the Design phase is the detailed design:

- Network diagrams (see [Network Topology Diagrams](#) for editable Microsoft Visio network drawings)
- Routing strategy

- Redundancy
- Call flows
- Traffic flows
- Equipment list
- Bill of materials

Major Concepts and Tasks in This Process

Go directly to the main design concepts and tasks:

- [Design Concepts](#)
- [Design Tasks](#)

Design Concepts

Read these conceptual, overview topics for the background knowledge you need to build an intelligent design.

- [Using SRND Documents](#)
- [Using Design Tools and Templates](#)

Using SRND Documents

Solution Reference Network Design (SRND) documents provide guidelines, recommendations, and best practices for implementing IP telephony network solutions. The following SRNDs are recommended for designing Cisco Collaboration Systems:

- [Cisco Collaboration Systems SRND Based on Cisco Unified Communications Manager 10.x](#)
- [Cisco Unified Communications SRND Based on Cisco Unified Communications Manager 9.x](#)
- [Cisco Unified Communications SRND Based on Cisco Unified Communications Manager 8.x](#)
- [Cisco Unified Communications SRND Based on Cisco Unified Communications Manager 7.x](#)
- [Cisco Unified Communications SRND Based on Cisco Unified Communications Manager 6.x](#)
- [Cisco Unified Contact Center Express SRND Release 10.0](#)
- [Cisco Unified Contact Center Express SRND Release 9.0](#)
- [Enterprise QoS System Reference Network Design](#)

**Note**

Additional SRND resources are available at http://www.cisco.com/en/US/netsol/ns742/networking_solutions_program_category_home.html.



Using Design Tools and Templates

Use these design tools to assist you in sizing your network:

- [Collaboration Sizing Tool](#) 

The Cisco Collaboration Sizing Tool assists users with hardware sizing of large or complex Collaboration solutions by calculating the call processing requirements for Collaboration products that have a major impact on performance and scalability.

- [IPC Tools](#) 

These tools, such as the IPC Resource Calculators, are intended to simplify and automate the process of sizing IP resources that are required for specific business operations. They are also useful for verifying and troubleshooting existing installations. The output from these tools can also be used as input to the [Cisco Unified Contact Center Express Configuration and Ordering Tool](#)  and the [Cisco Unified Communications Manager Capacity Tool](#). 

[Cisco Collaboration Sizing Tool](#)

A web-based tool that assists users with hardware sizing of large or complex Cisco Collaboration solutions by calculating the call processing requirements for products that have a major impact on performance and scalability. With the Cisco Collaboration Sizing Tool, system engineers with Cisco Collaboration solution experience or individuals with equivalent abilities can design and model solutions for existing and prospective customers. The tool requires various types of information to calculate the minimum size and type of devices required for a solution, such as the type and quantity of IP phones, gateways, and media resources. For most device types, the tool also requires the average number of call attempts per hour per device during the busy hour (known as busy hour call average or BHCA) and the average utilization time. The resulting calculations produced by the tool can be saved, copied, and sent to other users.

For additional information on design topics, see the documentation wiki (DocWiki) at http://docwiki.cisco.com/wiki/Unified_Communications_System_Design

Design Tasks

The following list is an overview of tasks for designing IP Telephony systems and is not meant to represent an ordered sequence of tasks.

- [Identify the Components That You Need](#)
- [Review Tested Site Models](#)
- [Review System Caveats](#)
- [Review System Test Results](#)
- [Develop Traffic Engineering Specifications](#)
- [Define Security Policies](#)
- [Design for High Availability](#)

Identify the Components That You Need

This topic provides links to descriptions of components, component documentation, and matrixes.

- [Collaboration Overview](#)- Describes the primary Cisco software components in Collaboration systems.
- [Review Release Matrix](#)—Identifies the components and their software versions that you need.
- [Component Resources Documentation for Collaboration](#)—Lists the component documentation.
- [Cisco Collaboration Compatibility Tool](#)—Provides tables that identify the compatible software release versions for each product in each Cisco Collaboration release.
- [Cisco Interoperability Portal](#)—Provides information about configuring Cisco technologies with third-party products and lists features that interoperate between Cisco solutions and other vendors.

The components that can be installed and configured in Collaboration environments are divided in categories as described in [Table 4-1](#).

Table 4-1 *Component Categories and Descriptions*

Category	Description
Call Control Components	<p>Call control components extend enterprise and midmarket telephony features and capabilities to packet telephony network devices such as IP phones, media processing devices, voice over IP (VoIP) gateways, and multimedia applications. Additional services, such as unified messaging, multimedia conferencing, collaborative contact centers, and interactive multimedia response systems, are made possible through open telephony APIs.</p> <p>Cisco Unified Communications Manager is an example of call control component.</p>
Server Application Components	<p>Server Application components are customized applications that help you streamline business processes and drive productivity through Internet Protocol (IP)-based Unified Communications. These components enable the rapid development, reliable execution and automated management of applications that integrate voice and video with enterprise and midmarket applications and data.</p> <p>Some examples of these components are Cisco Emergency Responder, Cisco Unified Attendant Consoles, and Cisco Unified Communications Manager IM and Presence Service.</p>
Conferencing Components	<p>Conferencing components provide integrated voice, video, and Web conferencing capabilities to enable remote meetings that are natural and effective with face-to-face quality, such as meetings, training sessions, and presentations.</p> <p>Some examples of these components are Cisco TelePresence Server and Cisco WebEx Meetings.</p>

Table 4-1 *Component Categories and Descriptions (continued)*

Category	Description
Voice Mail and Unified Messaging Components	<p>Voice mail and unified messaging components deliver powerful voice, integrated, and unified messaging options that transparently integrate with Microsoft Exchange, Lotus Domino, and Novell GroupWise. They scale to meet the needs of large, multisite organizations and offers extensive personalization options, a broad range of productivity enhancing features, and powerful migration tools, including:</p> <ul style="list-style-type: none"> • Interoperability with existing voice messaging and telephony systems • Custom keypad mapping of the telephony interfaces • Integrated context-sensitive help <p>An example of these components is Cisco Unity Connection.</p>
Endpoints and Clients Components	<p>Endpoint and client components integrate the management capabilities of IP-based networks with phones, pagers, and computers and use these for signaling, voice communications, and data communications.</p> <p>Some examples of these components are Cisco Desktop Collaboration Experience DX650 and Cisco TelePresence SX20 Quick Set.</p>
Security Components	<p>Security components are processes and technologies designed to quickly adapt to new threats to your network by using proactive, automated, real-time threat management.</p>
Network Management Components	<p>Network management components provide an integrated view of the entire Cisco Collaboration Systems and present the current operational status of each element of the IP communications network. In addition these components provide a low-cost, reliable method of monitoring and evaluating the quality of calls and diagnosing issues.</p> <p>Some examples of these components are Cisco Prime Collaboration Provisioning, Cisco Prime Collaboration Assurance, and Cisco Prime Collaboration Deployment.</p>
Communications Infrastructure Components	<p>Infrastructure components are network routers, switches, software, and other standards-based networking components that provide reliable connectivity that is more resilient and enables all the latest network services.</p>

Ordering Tools

- [Cisco Commerce Workspace \(CCW\)](#) — Cisco Commerce Workspace (CCW) provides functionality like Field Quoting, and reminder of Trade-In capability.

Review Tested Site Models

Cisco has developed a variety of site models as standard architectures. These models were tested and optimized for maximum efficiency and performance. You can derive your network design by choosing the deployment model that most closely matches your business and then adding the specific features and applications that meet your business needs. Review the following information:

- [IP Telephony Tested Deployments and Site Models](#)

Review System Caveats

System caveats describe unexpected behavior, defects, and product limitations discovered during system-level testing of Unified Communications components.

Check the latest release notes to make sure that your design has taken all system caveats into consideration. System caveats are listed under the [Limitations and Restrictions](#) section of the release notes.


Review System Test Results

System test results show the scope and extent of the testing conducted for Cisco Collaboration Systems in a test environment that is modeled on real-world deployments.

Depending on the network you are designing and your specific environment, use the system test results as a guide and supplement for your own site test and turn-up plan.

Develop Traffic Engineering Specifications

For background information on various traffic analysis concepts and features that are applicable to Voice over IP (VoIP), see [Traffic Analysis for Voice over IP](#). This document presents fundamental traffic theory, several statistical traffic models, application of traffic analysis to VoIP networks, and an end-to-end traffic analysis example.

Use the [Collaboration Sizing Tool](#)  to calculate the minimum number of active subscribers that are required to support a given installation. Inputs consist primarily of quantity and usage information on the various device types that are supported in a Cisco Unified Communications Manager system.

The following third-party traffic engineering tools are provided for your reference:

- [VoIP Bandwidth Calculator](#)
- [Online Erlang Traffic Calculators](#)

Define Security Policies

Refer to security policies in these guides:

- [Cisco Unified Communications Manager Security Guide](#)
- Specific component documents available in the [Component Resources Documentation for Collaboration](#) topic in the Resource Library.

Additional IP security information is included here for your reference:

- [Cisco Self-Defending Network](#)
- [Design Zone for Security](#)
- [Security Intelligence Operations](#)

Design for High Availability

Cisco Enterprise systems are designed for high availability. In order to achieve this, the design must include redundancy for failover and rapid recovery. For recommendations and design assistance from Cisco Advanced Services, see [Navigating the Road to Five Nines](#).

The Unified Communications Manager plays the key role in maintaining call processing following a failure in an Enterprise environment. This topic describes the following high-availability features that are built into Unified Communications Manager:

- [Unified Communications Manager Clusters](#)
- [Unified Communications Manager Redundancy Groups](#)
- [Keepalive Mechanism](#)



Note

The information in this topic applies to multiple Unified Communications Manager server clusters in a large enterprise environment and does not apply to Cisco Business Edition. Cisco Business Edition is a single-server solution designed for medium businesses up to a maximum of 500 employees and does not offer redundancy.

Unified Communications Manager Clusters

A cluster comprises a set of Unified Communications Manager servers (or *nodes*) that share the same database and resources. Unified Communications Manager servers can be configured to perform the following functions: database server, TFTP server, or application software server. You can dedicate a particular server to one function or combine several functions on one server, depending on the size of your network and the level of redundancy desired.

Each cluster can have only one database server (also called the *first node*) and usually one TFTP server (either separate or combined with another function). Cisco Systems recommends that large enterprise networks contain a dedicated Unified Communications Manager database server with other servers (called *subsequent nodes*) running the Unified Communications Manager application software. The Unified Communications Manager application software performs all call control, including signaling of endpoints, feature invocation, and calling restrictions. Large-scale networks typically use paired redundant application software servers, running in an active-active configuration, with endpoints evenly distributed across the two servers. The TFTP server provides configuration files for the endpoint devices and the associated firmware loads. Large enterprise networks typically use redundant TFTP servers.

Unified Communications Manager Redundancy Groups

A redundancy group comprises a prioritized list of up to three Unified Communications Manager servers. You can associate each group with one or more device pools to provide call processing redundancy. Each group must contain a primary Unified Communications Manager, and it may contain one or two backup Unified Communications Manager servers. If the primary Unified Communications Manager fails for any reason, the first backup Unified Communications Manager in the group takes control of the devices that were registered with the primary Unified Communications Manager. If you specify a second backup Unified Communications Manager for the group, it takes control of the devices if both the primary and the first backup Unified Communications Manager servers fail.

When a failed primary Unified Communications Manager comes back into service, it takes control of the group again, and the devices in that group automatically reregister with the primary Unified Communications Manager.

Keepalive Mechanism

A keepalive mechanism is an essential part of an Enterprise solution. Keepalives ensure that endpoints (typically phones and gateways) retain their communications path to a Unified Communications Manager server. Keepalives not only determine when the primary Unified Communications Manager server is no longer available, they also determine when the site has become completely isolated from a centralized call control system and must revert to some form of remote survivability capability such as Cisco Unified SRST. Keepalives avoid delays in establishing a call caused by searching for an available Unified Communications Manager server.

Additional Sites and Services

Steps to Success is a Cisco methodology that outlines the tasks required to complete a successful customer engagement. Registered users can visit the [Steps to Success](#) resource site for Cisco Unified Communications process flows.

Cisco Unified Communications Services is a Cisco service offering that provides engineering expertise and best practices.

- Registered users can visit the [Cisco Unified Communications Services](#) partner site.
- Nonregistered users can visit the [Services for Unified Communications](#) site.

For a solution that may fit your network design requirements, see Industry Solutions at <http://www.cisco.com/web/strategy/index.html>.



CHAPTER 5

Implement

Introduction to Implementation

The goal of implementation is to introduce the new system into the network with the least amount of disruption and the highest level of interoperability with the existing network. To minimize downtime, an essential component of this process is the implementation plan.



Tip

You can navigate to any topic on this tab by using the tab navigation pane at the left of the content pane. This navigation pane contains the table of contents (TOC) for the active tab.

Before You Begin

You should understand how to implement Cisco Unified Communications. For more information, see [Cisco Unified Communications Implementation](#).

Before you begin installing components, you should have a completed implementation plan from the detailed design. Use the equipment list and site specification from the detailed design to do the following:

- Order and stage equipment
- Perform detailed site survey
- Create site-specific installation guidelines

Your implementation plan should include:

- Deployment strategy
- Network maps and topology diagrams
- Installation and commissioning tests
- Site survey results
- List of all devices to be implemented
- Installation guidelines
- Configuration worksheets
- Test and turn-up plan

When You Are Done

All components are installed and ready to configure.

Major Tasks in This Process

- [Order Equipment](#)
- [Install and Configure System Components](#)
- [Conduct User Acceptance Test](#)
- [Preparing Your Network for Troubleshooting and Recovery](#)

Order Equipment

This topic includes links to ordering guides and tools that you need to choose your ordering options.


Quote Builder Tool

The Quote Builder tool is a solutions quoting application for Cisco Unified Communications products. Quote Builder is available to specialized partners and Cisco employees.

With Quote Builder, users can build a system quote with design documents to aid in the implementation of the solution. Quote Builder also validates designs for common deployments. Quote Builder generates a bill of materials, a network diagram, and design guides for deployment. To access Quote Builder, go to the following URL:

<http://www.cisco.com/web/partners/quotebuilder/index.html> 

Ordering Guides

[Ordering guides](#)  for most Cisco Unified Communications products are available for Cisco partners, Cisco sales staff, and Cisco service providers.

Install and Configure System Components

When implementing a Cisco Collaboration system, create a site-specific plan for your team. Describe what needs to be installed and configured. Your plan should list the referenced product-specific installation guides in [Table 5-1](#).

Your plan will help you manage timelines for implementing equipment and scheduling outages. Include an installation schedule, as well as a test plan that will verify that the operation conforms to the design objectives.

This section provides information about the components of the Cisco Collaboration Systems for IP Telephony Release 10.5(1) deployment. It does not describe installation procedures for individual components. For links to the complete documentation set for each system component, see the [Component Resources Documentation for Collaboration](#) topic on the Resource Library tab.

Performing Your System Installation

Before You Begin

See [Preparing for Your System Installation](#) on the Plan tab to plan your overall strategy.

Install IP Telephony Software Components

Once you have your installation plan and preparations in place, perform your system installation by following the guidelines and sequence in the individual product installation and configuration guides.

- See [Component Installation and Configuration Guides](#) for links to component installation and upgrade documentation.

For system configuration examples, as well as other system implementation topics, see the documentation wiki (docwiki) at http://docwiki.cisco.com/wiki/Unified_Communications_System_Implementation.

Component Installation and Configuration Guides

Using the steps from the documents listed in [Table 5-1](#), install or upgrade the required components for your network in the recommended order according to your site-specific implementation plan.

Table 5-1 *Component Installation and Configuration Documentation*

Product Category	Product Technology	Documentation Title
Call Control Components	Cisco Unified Communications Manager	<ul style="list-style-type: none"> • SAML SSO Deployment Guide for Cisco Unified Communication Applications • Installing Cisco Unified Communications Manager • Replacing a Single Server or Cluster for Cisco Unified Communications Manager • Upgrade Guide for Cisco Communications Manager • Cisco Unified Communications Manager Administration Guide • Cisco Unified Serviceability Administration Guide • Cisco Unified Communications Manager System Guide • Cisco Unified Communications Manager Bulk Administration Guide
	Cisco Business Edition 6000	<ul style="list-style-type: none"> • Installing Cisco Business Edition
	Cisco Business Edition 7000	<ul style="list-style-type: none"> • Cisco Business Edition 7000
	Cisco Unified Communications Manager Express	<ul style="list-style-type: none"> • Cisco Unified Communications Manager Express System Administrator Guide
	Cisco Unified Survivable Remote Site Telephony (SRST)	<ul style="list-style-type: none"> • Cisco Unified SRST System Administrator Guide • Cisco Unified SCCP and SIP SRST System Administrator Guide
	Cisco TelePresence Video Communication Server Cisco TelePresence Video Communication Server Expressway	<ul style="list-style-type: none"> • Configuration Guide for Cisco TelePresence Video Communication Server • Release Notes for Cisco TelePresence Video Communication Server • Troubleshooting Guide for Cisco TelePresence Video Communication Server

Table 5-1 Component Installation and Configuration Documentation (continued)

Product Category	Product Technology	Documentation Title
Contact Center Components	Cisco Unified Contact Center Express	<ul style="list-style-type: none"> • Installation Guide for Cisco Unified CCX and Cisco Unified IP IVR • Upgrade Guide for Cisco Unified CCX and Cisco Unified IP IVR • Getting Started with Cisco Unified Contact Center Express • Cisco CAD Installation Guide for CAD 8.0 for Unified CM • Cisco CAD Installation Guide for CAD 8.0 for Unified CME • Administration Guide for Cisco Unified CCX and Cisco Unified IP IVR • Cisco Unified Communications Operating System Administration Guide for Cisco Unified CCX and Cisco Unified IP IVR
	Cisco MediaSense	<ul style="list-style-type: none"> • Installation and Administration Guide for Cisco MediaSense • MediaSense User Guide
Server Application Components	Cisco Unified Communications Manager IM and Presence Service	<ul style="list-style-type: none"> • Installation and Upgrade Guide for Cisco Unified Presence • Deployment Guide for Cisco Unified Presence • User Guide for Cisco Unified Presence • Serviceability Configuration and Maintenance Guide for Cisco Unified Presence
	Cisco Emergency Responder	<ul style="list-style-type: none"> • Cisco Emergency Responder Administration Guide, “Installing Cisco Emergency Responder” chapter
	Cisco Unified Attendant Consoles	<ul style="list-style-type: none"> • Cisco Unified Attendant Consoles Installation Guides
Conferencing Components	Cisco TelePresence Server	<ul style="list-style-type: none"> • Cisco TelePresence Server 7010 Installation Guide • Cisco TelePresence Server 7010 and MSE 8710 v3.0 Locally Managed Mode Deployment Guide
	Cisco TelePresence MCU MSE Series	<ul style="list-style-type: none"> • Cisco TelePresence MCU MSE 8510 Installation Guide • Cisco TelePresence MCU MSE 8510 Creating and managing an MCU cluster
	Cisco TelePresence Conductor	<ul style="list-style-type: none"> • Install and Upgrade Guides
	Cisco WebEx Meetings Server	<ul style="list-style-type: none"> • Cisco WebEx Meetings Server Administration Guide • Cisco WebEx Meetings Server Planning Guide
	Cisco TelePresence Management Suite	<ul style="list-style-type: none"> • Install and Upgrade Guides
	Cisco WebEx Meeting Center	<ul style="list-style-type: none"> • User Guides

Table 5-1 Component Installation and Configuration Documentation (continued)

Product Category	Product Technology	Documentation Title
Voice Mail and Messaging Components	Cisco Unity Connection	<ul style="list-style-type: none"> • Installation Guide for Cisco Unity Connection • Upgrade Guide for Cisco Unity Connection • System Requirements for Cisco Unity Connection
Devices (Endpoints) Components	Cisco Unified SIP Phone 3900 Series	<ul style="list-style-type: none"> • Cisco Unified IP Phone 3900 Series Install and Upgrade Guides • Cisco Unified IP Phone 3900 Series End-User Guides • Cisco Unified IP Phone 3900 Series Maintain and Operate Guides
	Cisco Unified 69xx IP Phones	<ul style="list-style-type: none"> • Cisco Unified IP Phone 6900 Series End-User Guides • Cisco Unified IP Phone 6900 Series Maintain and Operate Guides
	Cisco IP Phone 7800 Series	<ul style="list-style-type: none"> • Accessibility Features for the Cisco IP Phone 7821, 7841, and 7861

Table 5-1 Component Installation and Configuration Documentation (continued)

Product Category	Product Technology	Documentation Title
Devices (Endpoints) continued	Cisco Unified 79xx IP Phones	<ul style="list-style-type: none"> • Cisco Unified IP Phone 7900 Series Install and Upgrade Guides • Cisco Unified IP Phone 7900 Series End-User Guides • Cisco Unified IP Phone 7900 Series Maintain and Operate Guides
	Cisco Unified 89xx IP Phones	<ul style="list-style-type: none"> • Cisco Unified IP Phone 8900 Series Install and Upgrade Guides • Cisco Unified IP Phone 8900 Series End-User Guides • Cisco Unified IP Phone 8900 Series Maintain and Operate Guides
	Cisco Unified 99xx IP Phones	<ul style="list-style-type: none"> • Cisco Unified IP Phone 9900 Series Install and Upgrade Guides • Cisco Unified IP Phone 9900 Series End-User Guides • Cisco Unified IP Phone 9900 Series Maintain and Operate Guides
	Cisco Desktop Collaboration Experience DX600 Series	<ul style="list-style-type: none"> • Regulatory Compliance and Safety Information for Cisco Desktop Collaboration Experience • Cisco Desktop Collaboration Experience DX600 Series Wireless LAN Deployment Guide
	Cisco TelePresence System EX Series	<ul style="list-style-type: none"> • Cisco TelePresence System EX60 Installation Sheet • Cisco TelePresence System EX90 Installation Sheet
	Cisco TelePresence Profile Series	<ul style="list-style-type: none"> • Cisco TelePresence System Profile 42-55-inch Installation Sheet
	Cisco TelePresence MX Series	<ul style="list-style-type: none"> • Install and Upgrade Guides
	Cisco TelePresence SX Series	<ul style="list-style-type: none"> • Install and Upgrade Guides
	Cisco TelePresence System Series	<ul style="list-style-type: none"> • Install and Upgrade Guides
	Cisco TelePresence Precision 60 Camera	<ul style="list-style-type: none"> • Cisco TelePresence Precision 60 Camera
	Cisco TelePresence SpeakerTrack 60	<ul style="list-style-type: none"> • Cisco TelePresence SpeakerTrack 60
Client Applications	Cisco Jabber for Android	<ul style="list-style-type: none"> • Install and Upgrade Guides
	Cisco Jabber iOS	<ul style="list-style-type: none"> • Install and Upgrade Guides
	Cisco Jabber for Windows	<ul style="list-style-type: none"> • Install and Upgrade Guides
	Cisco Jabber for Mac	<ul style="list-style-type: none"> • Install and Upgrade Guides
Network Management Components	Cisco Prime Collaboration	<ul style="list-style-type: none"> • Cisco Prime Collaboration Upgrade and Migration Guide, 10.5 • Cisco Prime Collaboration Quick Start Guide, 10.5
		<ul style="list-style-type: none"> • Cisco Prime Collaboration Quick Start Guide, 10.5
Licensing	Cisco Prime License Manager	<ul style="list-style-type: none"> • Software Downloads, Release and General Information

Table 5-1 *Component Installation and Configuration Documentation (continued)*

Product Category	Product Technology	Documentation Title
Communications Infrastructure Components	Cisco 2900 Series ISR	• Cisco 2900 Series Integrated Services Routers
	Cisco 3900 Series ISR	• Cisco 3900 Series Integrated Services Routers
	Cisco Unified Border Element	• Cisco Unified Border Element
	Cisco 4451-X ISR	• Cisco 4400 Series Integrated Services Routers
	Cisco IOS Software Release 15.1	• Cisco IOS 15.1M&T
	Cisco SRE 910 Service Module	• Cisco SRE 910 Service Module

Software Versions and System Caveats

For specific information on the product software versions used, system limitations, and known caveats, see the [System Release Notes for Cisco Collaboration Systems, Release 10.5.1](#).

Configuration Examples

Sample configuration commands for infrastructure components that are involved in the call flows are in downloadable zip files for all test bed components in [Configuration Command Files for Collaboration](#) in the Resource Library.

System-level configuration examples are available on the Cisco DocWiki at http://docwiki.cisco.com/wiki/Category:IP_Telephony_System_Configurations.

Introduction to Troubleshooting

This topic describes how to develop a system-level troubleshooting methodology as you install and configure a Cisco Unified Communications network for the first time. It also provides recommendations for preparing and documenting the network that may assist you in diagnosing and isolating problems when they occur. This topic contains the following sections:

- [System Troubleshooting Methodology](#)
- [Preparing Your Network for Troubleshooting and Recovery](#)

System Troubleshooting Methodology

The Implementation phase of your network deployment is an excellent time to develop a methodology for troubleshooting the network as a whole. Troubleshooting networking equipment at a system level requires solid detective skills. When a problem occurs, the list of potential suspects is long. You must collect detailed information and systematically narrow the list of potential causes to determine the root problem. This topic does not provide step-by-instructions for resolving problems that occur during network installation. Instead, this topic describes sound methods for troubleshooting your network using the following general steps:

1. [Gather Information on the Problem](#).
2. [Isolate Point\(s\) of Failure](#).
3. [Apply Tools to Determine the Problem's Root Cause](#).

Gather Information on the Problem

Problems are typically discovered and reported by one of the following types of users:

- External users trying to reach employees within your company
- External customers dialing into a call center to order products, obtain customer service, and so forth.
- Internal agents receiving incoming calls from a call queue or initiating outbound collection calls to customers.
- Internal users using administrative phones to call employees in other company locations or PSTN destinations, and perform basic actions such as call transfers and dialing into conferences.

As the network administrator, you must collect sufficient information from these users to allow you to isolate the problem. Detailed, accurate information will make this task easier. Table 5-2 lists recommended questions to ask users when they report a problem. As you turn up your network, you may consider putting these questions in an on-line form. A form will encourage users to provide more details about the problem and also put them into the habit of looking for particular error messages and indicators. Capturing the information electronically will also permit you to retrieve and re-examine this information in the future, should the problem repeat itself.

Table 5-2 Questions to Ask Users When They Report Problems

Ask this Question...	To Determine...
Did something fail or did it simply perform poorly?	Whether the issue relates to system degradation or a connectivity failure. An example of a failure is when a user dials a phone number and hears fast busy tone. An example of a performance problem is when a user dials into a conference call and hears “choppy” audio when other parties speak. Quality of service or performance issues require a different approach than connectivity or operational problems. You must still isolate the potential sources of the problem, but you will typically use performance management tools instead of log files.
What device were you trying to use?	The device type, model and version of software installed. It is also critical to capture the IP address assigned to the device, as well as its MAC address. In the case of IP phones, determining the phone’s active Cisco Unified Communications Manager server is also important. On Cisco Unified IP phones, these important network values can be displayed by pressing the Settings button and choosing the Network Configuration option from the menu.
Did it ever work?	If a device was recently installed and the problem occurred while making it work for the first time, or if the device was operating normally before the problem occurred. If the device was newly installed, the problem is most likely due to improper configuration or wiring of that particular device. Problems with devices that are already up and running can typically be traced back to one of two causes: (a) the user modifying their device, such as changing their configuration or upgrading software, or (b) a change or failure elsewhere in the network.
Exactly what action(s) did you perform?	The steps that led up to the problem, including which buttons were pressed and in which order. Capturing this information in detail is important so that you can consistently reproduce the problem.
What error message(s) appeared or announcements did you hear?	The visual and audio indicators of the problem. Ask users to provide the exact text that appears and any error codes in either an E-mail or on-line form. If the error indication was audible, ask the user to write down the announcement they heard, the last menu option they were able to successfully choose or the tone they heard when the call failed.

Table 5-2 Questions to Ask Users When They Report Problems (continued)

Ask this Question...	To Determine...
What time did the problem occur?	The date and time to compare against entries in log files. If the problem occurred on a Cisco Unified IP phone, make certain the user provides the timestamp that appears on their phone's display. Several Cisco components in a network may capture the same problem event in separate log files, with different ID values. In order to correlate log entries written by different components, you must compare the timestamps to find messages for the same event. Cisco Unified IP phones synchronize their date and time with their active Cisco Unified Communications Manager server. If all Cisco components in the network use Network Time Protocol (NTP) to synchronize with the same source, then the timestamps for the same problem messages will match in every log file.
What is the number of the phone you used and what was the phone number you called?	If the problem relates to a WAN or PTSN link, or a Cisco Unified Communications Manager dial plan issue. Ask the user the phone number he or she dialed (called number) and determine if the destination was within his or her site, another site within the corporate network, or a PSTN destination. Because the calling number (the number of the phone used) also affects call routing in some cases, capture this number as well.
Did you try to perform any special actions, such as a transfer, forward, call park, call pickup, or meet-me conference? Is the phone set up to automatically perform any of these actions?	If the problem is not directly related to the calling number or called number but rather to the supplementary service setup on Unified Communications Manager or the problem is at the destination phone the user tried to reach by transferring or forwarding the call.
Did you attempt the same action on another device?	If the problem is isolated to that user's device or represents a more widespread network problem. If the user cannot make a call from his or her phone, ask the user to place a call to the same destination using a phone in a nearby office.

Isolate Point(s) of Failure

After collecting information on the symptoms and behavior of the problem, to narrow the focus of your efforts you should:

- Identify the specific devices involved in the problem.
- Check the version of software running on each device.
- Determine if something has changed in the network.
- Verify the integrity of the IP network.

Identify Devices Involved in the Problem

In large- to medium-sized networks, it is crucial to identify the specific phones, routers, switches, servers and other devices that were involved in a reported problem. Isolating these devices allows you to rule out the vast majority of equipment within the network and focus your time and energy on suspect devices. To help you isolate which devices were involved in a problem, two types of information can prove invaluable:

- **Network topology diagrams:** It is strongly recommended that you have one or more diagrams that show the arrangement of all Cisco Unified Communications products in your network. These diagrams illustrate how these devices are connected and also capture each device's IP address and name (you may want to also have a spreadsheet or database of the latter information). This information can help you visualize the situation and focus on the devices that may be contributing to the reported problem. See [Network Topology Diagrams](#) for recommendations on how to prepare these diagrams.
- **Call flow diagrams:** Cisco equipment, including Unified Communications Manager servers, typically provide detailed debug and call trace log files. To interpret these log files, however, it is useful to understand the signaling that occurs between devices as calls are set up and disconnected. Using the network topology and call flow diagrams in conjunction with the log files, you can trace how far a call progressed before it failed and identify which device reported the problem. Examples of using call flow diagrams for problem isolation are shown in [Additional Sites and Services](#).

Check Software Release Versions for Compatibility

After you have identified which devices may be involved in the problem, verify that the version of software running on each device is compatible with the software running on every other device. As part of Cisco Unified Communications Release 10.5(1) verification, Cisco Systems has performed interoperability and load testing on simulated network environments running specific software versions. The [System Release Notes for Cisco Collaboration Systems, Release 10.5\(1\)](#) lists the combination of software releases that were tested.

However, if the combination of releases installed in your network does not match the values in the Release Matrix, it does not necessarily mean the combination is invalid. To check interoperability for a specific device and software release, locate and review its Release Notes. Release Notes contain up-to-date information on compatibility between the product and various releases of other products. This document also describes open caveats, known issues that may cause unexpected behavior. Before beginning extensive troubleshooting work, examine the Release Notes to determine if you are experiencing a known problem that has an available workaround.

**Tip**

The Bug Search requires that you are a Cisco partner or a registered Cisco.com user with a Cisco service contract. Using the Bug Search, you can find caveats for any release. To access the Bug Search, go to the <http://tools.cisco.com/Support/BugToolKit/>.

Determine if Network Changes Have Occurred

Before focusing on the particular device or site where the problem occurred, it may be useful to determine if a change was made to surrounding devices. If something has been added, reconfigured or removed from elsewhere in the network, that change may be the source of the problem. It is recommended that you track changes to the network such as:

- New phones added
- Modifications to Cisco Unified Communications Manager call routing settings, such as new directory numbers, route patterns and dial rules to support new sites or devices
- Changes to port configurations on switches, routers or gateways (new equipment, wiring changes or new port activation)
- Changes to IP addressing schemes (such as adding new subnets) that may have affected route tables

Verify the IP Network Integrity

Always remember that Cisco Unified Communications equipment relies on a backbone IP network. Many connectivity problems are not caused by configuration errors or operational failures on Cisco devices, but rather by the IP network that interconnects them. Problems such as poor voice quality are typically due to IP network congestion, while call failures between locations may be the result of network outages due to disconnected cables or improperly configured IP route tables.

Before assuming that call processing problems result from Cisco Unified Communications devices themselves, check the integrity of the backbone IP network. Keep the OSI model in mind as you perform these checks. Start from the bottom, at the physical layer, by checking that end-to-end cabling. Then verify the status of Layer 2 switches, looking for any port errors. Move from there to confirm that the Layer 3 routers are running and contain correct routing tables. Continue up the OSI stack to Layer 7, the application layer. To resolve problems occurring at the top levels of the stack, a protocol analyzer (or “sniffer”) may be useful. You can use sniffer to examine the IP traffic passing between devices and also decode the packets. Sniffers are particularly useful for troubleshooting errors between devices that communicate using Media Gateway Control Protocol (MGCP) or Session Initiation Protocol (SIP).

Apply Tools to Determine the Problem’s Root Cause

After you have eliminated the IP network as the source of the problem and you have isolated the specific Cisco Unified Communications components involved, you can start applying the many diagnostic tools provided by Cisco components.

[Table 5-3](#) lists the diagnostic tools and supporting troubleshooting documentation available for most components in an IP telephony network. Note that this summary table is provided for reference only. The procedures in [Troubleshooting Daily Operations](#) specify when to use each tool and provide links to the troubleshooting instructions in each component’s documentation where appropriate.

Table 5-3 IP Telephony Component Troubleshooting Tools and Documentation

Category	Component	Diagnostic Tools Available	Information Available In...
Call Control	Cisco Unified Communications Manager	Serviceability System tools: <ul style="list-style-type: none"> Alarms Real-Time Monitoring Tool window Trace log files <ul style="list-style-type: none"> Communications Manager trace log SDL trace log (under TAC direction) 	<ul style="list-style-type: none"> Troubleshooting Guide for Cisco Unified Communications Manager Cisco Unified Communications Manager Real-Time Monitoring Tool Administration Guide Cisco Unified Serviceability Administration Guide Cisco Unified CDR Analysis and Reporting Administration Guide Disaster Recovery System Administration Guide for Cisco Unified Communications Manager Troubleshooting TechNotes
	Cisco Business Edition	Serviceability System tools: <ul style="list-style-type: none"> Alarms Real-Time Monitoring Tool window Trace log files: <ul style="list-style-type: none"> Communications Manager trace log SDL trace log (under TAC direction) 	<ul style="list-style-type: none"> Troubleshooting Guide for Cisco Unified Communications Manager Cisco Business Edition Install and Upgrade Guides Cisco Unified Communications Manager Real-Time Monitoring Tool Guide
	Cisco Unified Communications Manager Express	IOS command line tools (such as Show commands and Debug trace utilities)	<ul style="list-style-type: none"> Troubleshooting Guides Troubleshooting TechNotes
	Cisco Unified Survivable Remote Site Telephony	IOS command line tools (such as Show commands and Debug trace utilities)	<ul style="list-style-type: none"> Cisco Unified SRST System Administrator Guide, “Monitoring and Maintaining Cisco Unified SRST” chapter
	Cisco TelePresence Video Communication Server (Cisco VCS)		<ul style="list-style-type: none"> Troubleshoot and Alerts

Table 5-3 IP Telephony Component Troubleshooting Tools and Documentation (continued)

Category	Component	Diagnostic Tools Available	Information Available In...
Contact Center	Cisco Unified Contact Center Express	Log files Alarms	<ul style="list-style-type: none"> • Cisco Unified Contact Center Express Operations Guide • Disaster Recovery System Administration Guide for Cisco Unified CCX and Cisco Unified IP IVR • Serviceability Administration Guide for Cisco Unified CCX and Cisco Unified IP IVR • Cisco CAD Troubleshooting Guide • Troubleshooting TechNotes
	Cisco MediaSense		<ul style="list-style-type: none"> • Cisco MediaSense Solution Reference Network Design guide • Troubleshooting Tips for Cisco MediaSense
Server Applications	Cisco Unified Communications Manager IM and Presence Service	System Configuration Troubleshooter Presence troubleshooter Microsoft Office Configuration Troubleshooter Trace log files Alarms	<ul style="list-style-type: none"> • Serviceability Configuration and Maintenance Guide for Cisco Unified Presence • Disaster Recovery System Administration Guide for Cisco Unified Presence • Troubleshooting TechNotes • Cisco Unified Communications Manager IM & Presence Service Documentation Roadmap
	Cisco Emergency Responder	E-mail alerts sent to administrator ERL Debug tool Cisco ER Admin Utility tool Call history logs SYSLOG collector tool that generates trace and debug files Windows Event Viewer	<ul style="list-style-type: none"> • Cisco Emergency Responder Administration Guide, “Troubleshooting Cisco Emergency Responder” chapter • Cisco Emergency Responder Troubleshooting Guide
	Cisco Unified Attendant Consoles		<ul style="list-style-type: none"> • Cisco Unified Attendant Console Troubleshooting Guide • Cisco Unified Attendant Consoles Troubleshooting TechNotes

Table 5-3 IP Telephony Component Troubleshooting Tools and Documentation (continued)

Category	Component	Diagnostic Tools Available	Information Available In...
Conferencing	Cisco TelePresence Server	Troubleshooting Guides Security Advisories, Responses and Notices	<ul style="list-style-type: none"> • Cisco TelePresence Server Troubleshooting Guides • Cisco TelePresence Server Security Advisories, Responses and Notices
	Cisco TelePresence MCU MSE Series		<ul style="list-style-type: none"> • Cisco TelePresence MCU MSE Series Troubleshooting Guides
	Cisco TelePresence Conductor		<ul style="list-style-type: none"> • Troubleshoot and Alerts
	Cisco Telepresence Management Suite		<ul style="list-style-type: none"> • Troubleshoot and Alerts
	Cisco WebEx Meetings Server		<ul style="list-style-type: none"> • Cisco WebEx Meetings Server Troubleshooting Guides • Cisco WebEx Meetings Server Security Advisories, Responses and Notices
	Cisco WebEx Meeting Center		<ul style="list-style-type: none"> • Troubleshoot and Alerts
Voice Mail and Unified Messaging	Cisco Unity Connection	Serviceability System tools: <ul style="list-style-type: none"> • Alarms • Real-Time Monitoring Tool window Cisco Unity Diagnostic Tool (UDT): <ul style="list-style-type: none"> • Macro trace logs • Micro trace logs CuVrt service verbose logging	<ul style="list-style-type: none"> • Troubleshooting Guide for Cisco Unity Connection • Cisco Unified Serviceability Administration Guide • Cisco Unified Real-Time Monitoring Tool Administration Guide • Troubleshooting TechNotes

Table 5-3 IP Telephony Component Troubleshooting Tools and Documentation (continued)



Category	Component	Diagnostic Tools Available	Information Available In...
Devices (Endpoints)	Cisco Unified IP Phones	Network configuration, status and phone model information on Settings menu	<ul style="list-style-type: none"> • End-User Guides <ul style="list-style-type: none"> – Cisco Unified SIP Phone 3900 Series – Cisco Unified IP Phone 6900 Series – Cisco IP Phone 7800 Series – Cisco Unified IP Phone 7900 Series – Cisco Unified IP Phone 8900 Series – Cisco Unified IP Phone 9900 Series • Cisco Unified IP Phone Administration Guides for Cisco Unified Communications Manager, “Troubleshooting and Maintenance” chapters <ul style="list-style-type: none"> – Cisco Unified SIP Phone 3900 Series – Cisco Unified IP Phone 6900 Series – Cisco IP Phone 7800 Series – Cisco Unified IP Phone 7900 Series – Cisco Unified IP Phone 8900 Series – Cisco Unified IP Phone 9900 Series • Error Message Decoder  • Output Interpreter  • Troubleshooting TechNotes <ul style="list-style-type: none"> – Cisco Unified IP Phone 7900 Series – Cisco Unified IP Phone 8900 Series – Cisco Unified IP Phone 9900 Series • Error Message Decoder  • Output Interpreter 
	Cisco Desktop Collaboration Experience DX650		<ul style="list-style-type: none"> • Cisco Desktop Collaboration Experience DX600 Series Administration Guide, Release 10.1(2) • Cisco Desktop Collaboration Experience DX650 User Guide, Release 10.1(2) • Cisco Desktop Collaboration Experience DX Series Translated End-User Guides

Table 5-3 IP Telephony Component Troubleshooting Tools and Documentation (continued)

Category	Component	Diagnostic Tools Available	Information Available In...
Devices (Endpoints) continued	Cisco TelePresence SX Series		<ul style="list-style-type: none"> • Troubleshoot and Alerts
	Cisco TelePresence MX Series		<ul style="list-style-type: none"> • Troubleshoot and Alerts
	Cisco TelePresence System EX	Log files Release Notes Troubleshooting Guides	<ul style="list-style-type: none"> • Cisco TelePresence System EX60 and EX90 Administrator Guide • Cisco Security Advisory: Default Credentials for Root Account on Tandberg E, EX and C Series • Cisco TelePresence System Profiles using Codec xxx Administrator Guide
Client Applications	Cisco Jabber for Android		<ul style="list-style-type: none"> • Maintain and Operate
	Cisco Jabber for iOS		<ul style="list-style-type: none"> • Maintain and Operate
	Cisco Jabber for Mac		<ul style="list-style-type: none"> • Troubleshoot and Alerts
	Cisco Jabber for Windows		<ul style="list-style-type: none"> • Troubleshoot and Alerts
Network Management	Cisco Prime Collaboration		<ul style="list-style-type: none"> • Troubleshooting Cisco Prime Collaboration • Cisco Prime Collaboration Troubleshooting TechNotes
Licensing	Cisco Prime License Manager		<ul style="list-style-type: none"> • Cisco Prime License Manager Introduction
Communications Infrastructure	Cisco IOS Software Release 15.1		<ul style="list-style-type: none"> • Cisco IOS Software Release 15.1 Troubleshooting TechNotes
	Cisco 3900 Series Integrated Services Router		<ul style="list-style-type: none"> • Troubleshoot and Alerts
	Cisco 2900 Series Integrated Services Router	IOS command line tools (such as Show commands and Debug trace utilities)	<ul style="list-style-type: none"> • Cisco 3900 Series, 2900 Series, and 1900 Series Software Configuration Guide, “Using ROM Monitor” chapter
	Cisco 4400 Series Integrated Services Router	IOS command line tools (such as Show commands and Debug trace utilities)	<ul style="list-style-type: none"> • Troubleshooting Guide for the Cisco 4451-X Integrated Services Router
	Cisco Unified Border Element		<ul style="list-style-type: none"> • Cisco Unified Border Element Troubleshooting TechNotes

Preparing Your Network for Troubleshooting and Recovery

Before your network becomes operational, you can take several proactive steps to make troubleshooting easier, including:

- Produce network topology diagrams to help you isolate potential sources of problems.
- Synchronize the date and time on all servers.
- Set trace/logging levels on key devices so that diagnostic information is available when problems occur.

Network Topology Diagrams

One of the first lines of defense is possessing current topology information. One of the most important pieces of topology information is a detailed network diagram (usually created using Microsoft Visio or a similar application). At a minimum, your network topology diagrams should include the following information:

- The name assigned to each major device (typically the DNS name)
- IP addresses for all devices in the network
 - Addresses for each router, core and access switch
 - Addresses for all telephony and application servers, including the IP address for each server in a Cisco Unified Communications Manager cluster
 - DHCP address range for addresses assigned to endpoints such as IP phones and agent workstations
- Phone extension number ranges assigned to sets of agents or users, as well as the main inbound dial-up numbers for each location. This information is useful in resolving dial plan configuration errors.
- WAN IP and PSTN links between sites.

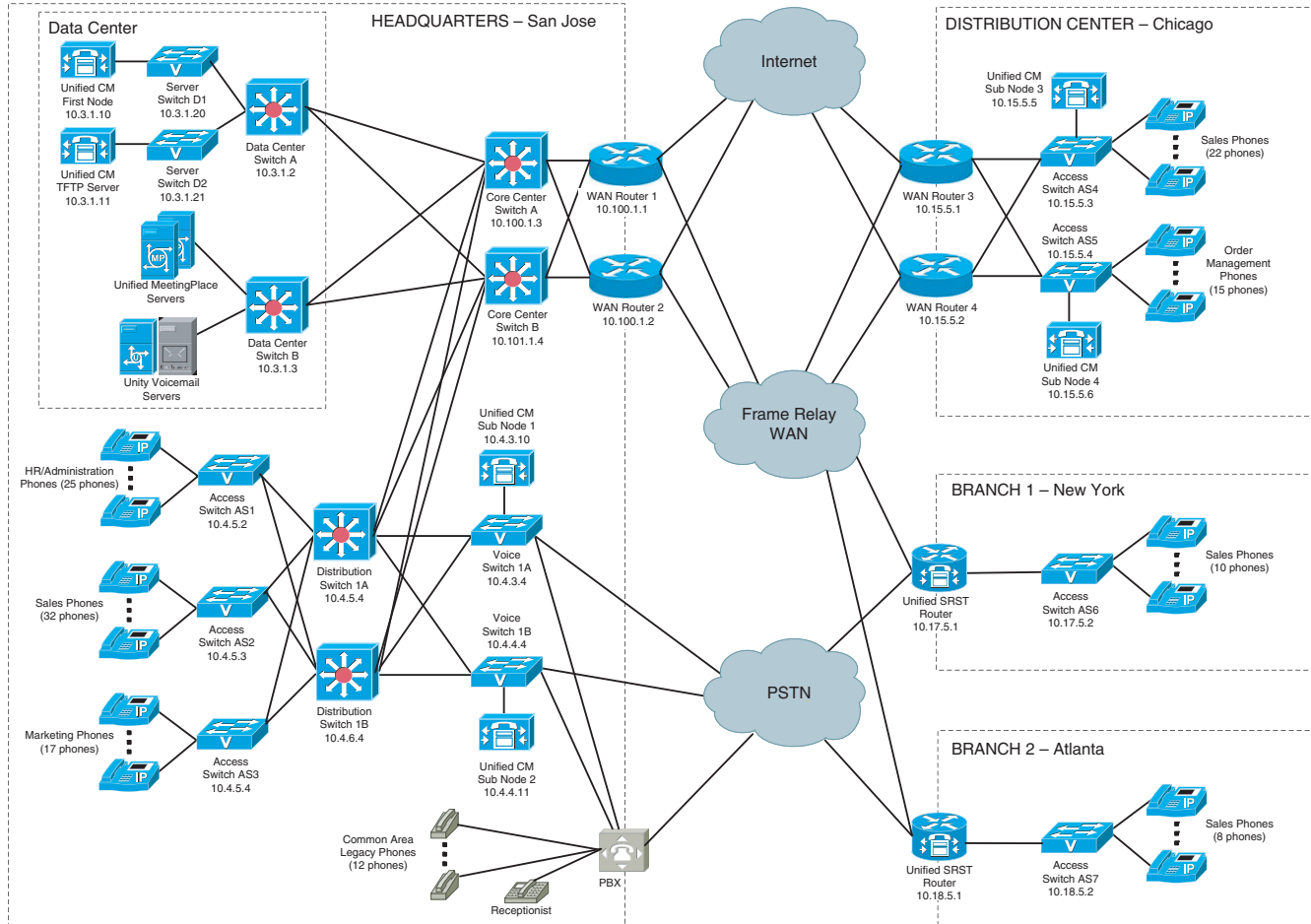
This information is critical for isolating which components are involved in a particular problem. For medium- to large-sized networks, you may want to take a “layered” approach in your diagrams. Create a high-level diagram that illustrates the overall physical layout of your network, including all sites and the links between them. Then for each site create additional diagrams that show detailed addressing information, port numbers and dial plan configurations.

**Tip**

Frequent adds, changes and upgrades to your network can quickly make these diagrams out-of-date. Inaccurate diagrams slow down the troubleshooting process and may lead to misdiagnosing the problem. Remember to keep these diagrams as current as possible.

Figure 5-1 shows a typical high-level topology diagram for a medium-sized enterprise IP telephony network. Note that only device names and IP addresses are listed in the diagram.

Figure 5-1 IP Telephony Network Topology Diagram Example



Synchronizing Server Date and Time

The best resources for diagnosing problems within your network are the debug and trace log files produced by individual Cisco devices. Tracing can be enabled on multiple devices and the log file output compared to isolate problems. In order to correlate messages for the same activity in different log files, you must compare the message timestamps and the source device MAC and IP addresses (there is no universal call ID value shared between Cisco devices). You should synchronize every device to the same date and time source so that the timestamps match. To accomplish this synchronization, set each device to obtain its date and time from the same Network Time Protocol (NTP) source.

For Cisco IOS-based devices (switches, routers or voice gateways), you can configure each device to act as a NTP client and periodically poll a master NTP source using the following command:

```
ntp server ip-address [version number] [key keyid] [source interface] [prefer]
```

Additional IOS commands are available to establish a device as a NTP peer (operating as the master source for other devices), as well as setting up NTP broadcasting instead of polling. See the [Cisco IOS Configuration Fundamentals Command Reference](#) for details on these IOS commands.

Recommended Trace/Logging Settings

In order to have diagnostic information available when you begin to research problems, you must configure devices in your network to capture signaling, processing and other activity in log files.

Cisco Unified Communications Manager Trace Settings

Trace settings for Cisco Unified Communications Manager servers are maintained using the Cisco Communications Manager Serviceability graphical interface. There are two ways to set trace logging levels for Unified Communications Manager services:

- **Customize trace levels for individual parameters:** This approach offers a high-degree of control and flexibility over the trace output. However, in order to use this approach you should understand not only the significance of each parameter, but also the impact of tracing on Unified Communications Manager server performance. For example, setting trace levels to “Error” has a minimal impact to CPU cycles while leaving the “Detail” level set for long periods of time may impact call processing. For instructions on setting individual trace levels, see the [Cisco Unified Serviceability Administration Guide, “Configure Trace” chapter](#).
- **Apply predefined trace levels:** This approach allows you to quickly enable and disable tracing for each Unified Communications Manager service based on predefined levels. You can also use these default troubleshooting trace settings in combination with customized settings to temporarily override the your custom settings. For instructions on using the Troubleshooting Trace Settings option in the Cisco Unified Communications Manager Serviceability interface, see the [Cisco Unified Serviceability Administration Guide, “Trace Setting Troubleshooting” chapter](#).

Conduct User Acceptance Test

After the components are configured and integrated with other IP Telephony applications, the field engineer prepares the system for the user acceptance test. Test scripts are run and compared against expected results. Any variability in network performance is noted and addressed before the user acceptance test.

Testing the customer solution involves the following tasks:


- Determine the user acceptance test parameters and deliverables and record these in the user acceptance test plan.
- Conduct a prelaunch test—Using an incremental approach, test the solution against the system design in a low-risk environment with limited users. If the system is stable, the rollout pace is increased until the full implementation is operational.
- Network ready for use acceptance—The customer signs the Ready-for-Use Acceptance Letter acknowledging that the acceptance test yielded satisfactory results.

Train End Users

The final stage of the Implement phase is helping to ensure that the customer system administration team and end users are trained to take over management of the new system.

Cisco Systems offers several training and certification programs for customers to maximize the usage of their newly adopted systems.

Additional Sites and Services

Steps to Success is a Cisco methodology that outlines the tasks required to complete a successful customer engagement. Registered users can visit the [Steps to Success](#)  resource site for Cisco Unified Communications process flows.

Cisco Unified Communications Services is a Cisco service offering that provides engineering expertise and best practices.

- Registered users can visit the [Sell and Market Services](#)  partner site.
- Nonregistered users can visit the [Services for Unified Communications](#) site.



CHAPTER 6

Operate

Introduction to Operating the System

To ensure that your network operates efficiently and reliably, your daily operations should consist of system and performance management practices. These practices include scheduled routine maintenance; keeping maintenance records; and maintaining up-to-date upgrade, troubleshooting, and recovery strategies.



Tip

You can navigate to any topic on this tab by using the tab navigation pane at the left of the content pane. This navigation pane contains the table of contents (TOC) for the active tab.

Before You Begin

User acceptance testing is completed and any problems that surfaced have been resolved. Users have been trained in using the new system.

Output of This Process

The Operate phase produces data that will inform the Optimize phase.

Major Tasks in This Process

- [Managing Your System](#)
- [Backing Up and Restoring Components](#)
- [Using Network Monitoring Tools](#)
- [Using Managed Services Documents](#)
- [Troubleshooting Daily Operations](#)

Managing Your System

This topic provides a high-level summary of the ongoing tasks required for managing your system and the options for how these tasks can be performed. For detailed maintenance and operation guides for each component in your Cisco Collaboration Systems, see the product documentation listed in [Component Resources Documentation for Collaboration](#) in the Resource Library.

System Management Tasks

Managing a Cisco Collaboration System consists of performing the following activities:

- Integrating monitoring and management tools—Select, order, configure, integrate, and test a set of tools for monitoring and managing the Cisco Collaboration Systems.
- Monitoring—Set thresholds, monitor events, and generate notifications when service-impacting events occur.
- Ticketing—Generate and track system trouble tickets for each event.
- Diagnosing incidents—Analyze and troubleshoot incidents to determine the cause.
- Resolving incidents—Define and execute an action plan which can include performing break and fix activities, applying software updates and patches, managing hardware replacements, and executing change management processes.
- Managing changes in the network—Define a change management process for performing moves, adds, changes, and disconnects (MACDs) for your Cisco Collaboration Systems including network devices, phones/endpoints, software upgrades, voice-mail boxes, dial plan updates, security patches, OS applications, and voice applications.
- Archiving configurations—Back up device configurations daily and restore device configurations when necessary.
- Managing voice as a network service—Track, measure and resolve quality of service (QoS) issues such as jitter, delay, and dropped packets, and monitor service level agreements (SLAs) with service providers.
- Managing security posture—Detect, analyze, and address security events.
- Reporting—Define, develop, and generate performance, availability, event, and inventory reports.
- Backing up and restoring system components—Define backup methodologies and schedules, define a verification process for backups, secure storage of backups, and document backup processes.

System Management Options

There are two options for managing a Cisco Collaboration System:

- Do It Yourself—In this model, you are responsible for managing the entire Cisco Collaboration System. This approach requires developing business processes; integrating, provisioning and maintaining network management tools; and developing data and voice management skills and knowledge. Cisco offers tools as a means for monitoring your network; see [Using Network Monitoring Tools](#) for more information.
- Out tasking Hybrid Model—Using the [Services Portfolio](#) approach, Cisco and its partners provide a broad portfolio of end-to-end services and support that can help increase your Cisco Collaboration system's business value and return on investment. This approach includes two services that provide different levels of management:
 - [Cisco Unified Communications Essential Operate Service](#) —Cisco Essential Operate Services, part of Product Support in the Cisco Services Lifecycle framework, is available for Unified Communications software, collaboration software, and network management software. It provides the support you need to help reduce downtime, improve performance, and lead to higher productivity.
 - [Cisco Unified Communications Remote Management Service](#) includes monitoring and reporting plus managing day-to-day system issues such logical moves, adds, changes, and disconnects; resolving incidents; performing configuration backups; and reporting.

For more information about the Cisco Unified Communications Essential Operate Service, Cisco Unified Communications Remote Management Service or other Cisco Unified Communications services, see <http://www.cisco.com/en/US/products/sw/voicesw/services.html> or contact your Cisco service account manager.

Backing Up and Restoring Components

This topic provides details on backup and restore for Cisco Collaboration components. First and foremost, the backup of Cisco Collaboration components needs to be incorporated into your corporate-wide backup operations. It is an important aspect of disaster recovery and is also essential before doing component upgrades. If you do not have a process in place, you must develop and document a backup and recovery management process. Some items to consider for this process are the following:

- Provide proper storage of operating system and Cisco Collaboration application CDs.
- Define incremental and full backup methodologies and schedules, assign an owner for each Collaboration component and database server.
- Define a verification process for backups:
 - Monitor backup logs on a daily basis for errors.
 - Periodically restore backup images to ensure validity.
- Secure onsite and offsite storage of backups.
- Develop well documented processes for system and configuration restoration.
- Ideally, provide central location(s) (for example, SFTP servers) for backup of data from all the Cisco Collaboration components.

The following topics provide backup and restore details on a component basis along with links to the appropriate component documentation:

- [Cisco Unified Communications Manager](#)
- [Cisco Unified Communications Manager Express](#)
- [Cisco Unified Contact Center Express](#)
- [Cisco Unified Communications Manager IM and Presence Service](#)

For additional information on backing up and restoring Unified Communications system components, as well as other system operations topics, see the documentation wiki (DocWiki) at http://docwiki.cisco.com/wiki/Unified_Communications_System_Operations.

Cisco Unified Communications Manager

Cisco Unified Communications Manager provides the Disaster Recovery System (DRS) for full backup and restore for all servers in a Unified Communications Manager cluster. The DRS performs a cluster-level backup, which means that it collects backups for all servers in a Unified Communications Manager cluster to a central location and archives the backup data to a physical storage device (tape or SFTP). For customers with multiple clusters, DRS must be configured per cluster.

DRS is invoked via the Unified Communications Manager Platform Administration. It allows you to perform scheduled (daily, weekly, monthly) automatic or user-invoked backups. DRS only supports a single backup schedule at a time. It provides a history (last 20 operations) of backup and restore operations.

With Cisco Unified Communications Manager Business Edition, DRS will also provide backup and restore capabilities for Unity Connection.

**Note**

DRS does not support hostname or IP address change during restore. For more information about the Disaster Recovery System, see the [Disaster Recovery System Administration Guide for Unified Communications Manager](#).

Cisco Unified Communications Manager Express

Cisco Unified Communications Manager Express is an application that runs on Cisco IOS. Therefore, for backup and restore, the IOS facility for backup and restoring configuration data would be used. For more information, see specific documents in [Cisco Unified Communications Manager Express Documentation](#).

Cisco Unified Contact Center Express

For Unified CCX Release 10.5(1), the Backup and Restore application is described in Cisco Unified Contact Center Express Disaster Recovery System Administration Guide. For more information, refer [Cisco Unified Contact Center Express Configuration Guides](#).

Cisco Unified Communications Manager IM and Presence Service

Cisco Unified Communication Manager IM and Presence Service will use the Disaster Recovery System (DRS) for full data backup and restore capabilities of all administration. For more information, see the [Disaster Recovery System Administration Guide for Cisco Unified Communications Manager](#).

Using Network Monitoring Tools

Cisco Prime Collaboration allows businesses to actively monitor their Cisco Unified Communications solution to discover potential problems, maintain quality and user satisfaction, and help minimize service downtime. The following network monitoring tool is primarily used:

- [Cisco Prime Collaboration](#)

For more information about network monitoring, as well as other system operations topics, see the Cisco Unified Communications category on the documentation wiki (DocWiki) at http://docwiki.cisco.com/wiki/Cisco_Unified_Communications.

Cisco Prime Collaboration

Collaboration provides simplified, unified management for Cisco voice and video networks that allows enterprises to extract the full value from their unified communications and collaboration technology investments. The solution removes management complexity and provides automated, accelerated provisioning, real-time monitoring, proactive troubleshooting, and long-term trending and analytics in one integrated product. A premier operations experience is provided through an intuitive user interface

and an optimized operator methodology including automated workflows that ease implementation and ongoing administration. For a detailed overview of the Cisco Prime Collaboration product and literature, see [Cisco Prime Collaboration Support Page](#).

Cisco Prime Collaboration is a comprehensive video and voice service assurance and management system with a set of monitoring, reporting, and diagnostic capabilities that help ensure end users receive a consistent, high-quality video and voice collaboration experience.

Prime Collaboration offers the following benefits:

- Provides provisioning, monitoring, reporting, and diagnostic capabilities for Unified Communications Systems.
- Supports timely, end-to-end visibility and isolates voice and video-related issues for sessions (calls), endpoints, and the network.
- Reduces time to troubleshoot and recover from service-affecting problems.
- Provides detailed analysis of the media (voice and video) path with critical fault and performance statistics that enable you to isolate network devices that cause service degradation.
- Delivers reports that enable you to track usage and problem history.

For in-depth details on using Cisco Prime Collaboration for networking monitoring, see

- [Cisco Prime Collaboration 10.5 Documentation Overview](#)

Using Managed Services Documents

Managed Services documents provide essential information to ensure the continuous operation of a Cisco Collaboration System by monitoring faults, configuration, performance, and security. These documents define a component's Management Information Base (MIB) structure and also explain the syslogs, alerts, and alarms elements that service providers can incorporate in their managed services. They also describe the component's system architecture while outlining basic concepts such as Simple Network Management Protocol (SNMP) and presenting the capabilities of the management tools available for that component.

Refer to the individual product pages for more information on the managed services.

Troubleshooting Daily Operations

This topic describes how to diagnose and resolve system-level problems that occur during daily operations of a Cisco Collaboration network. It contains the following sections:

- [Common Problems Reported by Users](#)
- [PBX Interoperability Issues with Cisco Unified Communications Manager](#)
- [Using Call Flows to Resolve Call Processing Problems](#)
- [Failover and Recovery Procedures](#)

For an expanded list of general problem areas, as well as other system troubleshooting topics, see the documentation wiki (DocWiki) at

http://docwiki.cisco.com/wiki/Unified_Communications_System_Troubleshooting.

Common Problems Reported by Users


This section describes basic approaches to diagnose and resolve common problems reported by end users. Note that this section demonstrates various tools and diagnostic approaches available in the context of specific problems, but does not provide a comprehensive list of all possible problems that may occur. Problems described in this section include:

- [One-Way Audio](#)
- [Poor Voice Quality](#)

One-Way Audio

One-way audio and no audio at all (no-way audio) are problems that are fairly common during a new network installation. The majority of these problems are caused by misconfigurations. For one-way audio problems, always pay attention the direction in which the one-way audio is occurring. For no audio in either direction, the troubleshooting methodology is the same. You might need to repeat the procedure for each direction of audio, but more likely you will find the source of the problem when trying to troubleshoot one direction. There are several steps you can take to troubleshoot a one-way/no-way audio problem:

1. [Verify Bidirectional IP Connectivity](#).
2. [Check Cisco IOS Software Gateway Configurations](#).
3. [Check for NAT or Firewall Restrictions](#).

For additional directions on troubleshooting one-way audio problems, refer to the [Troubleshooting One-Way Voice Issues Tech Note](#) .

Verify Bidirectional IP Connectivity

You should verify IP connectivity as the first step in troubleshooting a one-way or no-way audio problem because IP connectivity must be present for voice packets to be exchanged between two devices. A large number of one-way or no-way audio problems are caused by lack of IP connectivity. Check that:

- If the two endpoints involved in the call are on different IP subnets, each endpoint has the correct default gateway and subnet mask settings
- If one of the endpoints is a Unified IP phone, the DHCP scope has an incorrectly configured default gateway parameter.
- If one of the endpoints is a Cisco IOS software gateway, the default route is correct. Also, ping the other endpoint from the gateway. If the ping is successful, you know that you have IP connectivity. If the ping is unsuccessful, perform a **tracert** to determine where the problem lies.



Note

Remember that signaling packet traffic is always between Unified Communications Manager and the endpoint, whereas the RTP voice packet traffic is directly between the endpoints. So just because the endpoints are registered to Unified Communications Manager and can set up a call through Unified Communications Manager does not mean that the endpoints have proper IP connectivity between them.

Another useful tool for troubleshooting such a problem is the help (i or ?) button on Cisco Unified IP phones. Press the help (i or ?) button twice in quick succession during an active call. The display shows you receive and transmit statistics for the call. If you do not see the receive counter (RxCnt) incrementing, the packets are probably not arriving on that IP phone. If you go to the originating IP

phone and the transmit count (TxCnt) is incrementing, the packets are probably being lost somewhere in the network. If a ping or traceroute does not provide enough information about where the packets are being lost, you may need to connect a sniffer to the network and perform the following steps:

1. Connect the sniffer to the back of the originating IP phone and make verify that the phone is actually transmitting packets.
2. On the originating phone, verify that the IP address and MAC address information is correct.
3. If the network settings on the originating phone are correct, go to the terminating IP phone to verify that the packets are not arriving.
4. If the voice packets are not arriving at the terminating phone, move the sniffer from network hop to network hop to isolate where the packets are being dropped. A common reason for a problem such as this is a missing or improperly configured IP route.

Check Cisco IOS Software Gateway Configurations

There are various reasons why you might encounter one-way audio on calls to a Cisco IOS software gateway. Most of these problems can be solved using simple configuration commands.

1. Check if IP routing is enabled on the gateway that you are using—You do not need to be running a routing protocol such as RIP, EIGRP, or OSPF, but IP routing must not be disabled. Make sure that the **no ip routing** command is not in your configuration. If it is, be sure to eliminate it by configuring the **ip routing** command. You can also issue the **show ip route** command to see if IP routing is enabled. If IP routing is disabled, there are no routes listed in the output, and the list of routing protocols is not present.
2. Determine if the VoIP subsystem is enabled—The VoIP subsystem in Cisco IOS software uses the IP routing code to aid in encapsulating and transmitting the VoIP packets, so the subsystem must be enabled to transmit and receive VoIP packets. It does not need the IP routing code to perform signaling such as H.323 or MGCP, so the signaling still works with IP routing disabled.
3. Check IP address configurations on gateway interfaces—Another common occurrence of one-way audio appears on Cisco IOS software H.323 voice gateways that have more than one data interface, such as a gateway that has both an Ethernet connection to the LAN and a serial connection to the WAN. When an H.323 gateway is configured in Cisco Unified Communications Manager Administration, you configure a specific IP address. Cisco Unified Communications Manager always uses this IP address for all its signaling to the gateway; however, Cisco IOS software voice gateways by default use the IP address of the interface that is closest to the destination. This could be a problem if Unified Communications Manager is connected via one interface and the device to which the RTP audio stream is destined for is connected to a different interface. To force the voice gateway to always use the same IP address, configure the **h323-gateway voip bind srcaddr ip-address** command on the interface that you are using for signaling on the Cisco IOS software voice gateway. Make sure this is the same IP address configured in Cisco Unified Communications Manager Administration. Failure to do so could result in one-way audio when the gateway tries to use a different source interface than the one configured in Unified Communications Manager.
4. Configure voice rtp send-recv on the gateway—Sometimes you have one-way audio problems only when calling specific numbers, such as 411 or 911 in the North American numbering plan (NANP) or after you transfer a call or put it on hold. If you are having these problems when going through a Cisco IOS software voice gateway, be sure that the **voice rtp send-recv** command is configured on the gateway. Numbers such as 411 and 911 sometimes do not send back answer supervision (that is, an ISDN connect message) when the remote end answers. As a result, the Cisco IOS software voice gateway does not cut through audio in both directions to prevent toll fraud. Configuring the **voice rtp send-recv** command forces the voice gateway to cut through audio in both directions immediately.

5. If you are using a Cisco AS5350 or AS5400 as a gateway, configure the **no voice-fastpath enable** command in global configuration mode—When enabled, this command causes the voice gateway to cache the IP address and UDP port number information for the logical channel opened for a specific call and forwards the packets using the cached information. This helps marginally reduce CPU utilization in high-call-volume scenarios. Because of how Cisco Unified Communications Manager opens and closes logical channels to redirect RTP audio streams, such as in the case of a transfer or music on hold (MOH) server, the Cisco AS5350 and AS5400 cache the IP address information of the old IP address. Therefore, you end up with one-way audio when the call gets redirected to a new IP address because the voice gateway still uses the cached information instead of the newly negotiated information.

Check for NAT or Firewall Restrictions

One common cause of one-way or no-way audio is when Network Address Translation (NAT), Port Address Translation (PAT), or firewalls exist between two endpoints. The SCCP protocol embeds IP addresses in the IP packet's payload to signal which IP address to send RTP packets to. If the device performing NAT or PAT is unaware of this fact, the embedded IP addresses are not translated. Therefore, one-way or no-way audio results.

Firewalls can also be a problem if they are unaware of the voice traffic passing through them. Firewalls often are configured to block all UDP traffic going through them. Because voice traffic is carried over UDP, it might be blocked while the signaling carried over TCP is passed. A sniffer is the best tool for debugging such a scenario. If both devices appear to be transmitting audio but the audio is not reaching the opposite side, take a sniffer trace at each hop along the way until you find the hop where the audio is not passing through. If the firewall is blocking UDP packets, you might need to open a hole in it to allow the voice traffic to pass through.

Problems Occurring After the Call Connects Successfully

The scenarios discussed so far are cases in which you have one-way audio or no-way audio from the beginning of the call or after a hold/transfer. Occasionally, however, you might encounter scenarios in which a call is up and suddenly becomes one-way or audio disappears entirely. Network problems are largely to blame for failures of this sort. Ensure that network connectivity between the two endpoints still exists and that nothing on the network might be causing intermittent network connectivity. An example would be a *flapping* network connection—a network connection that is transitioning between up and down states over and over again—or a routing protocol that cannot converge correctly. Again, a sniffer is the best tool for diagnosing this kind of problem. The best place to start is on the device that originates the RTP stream to ensure that the stream is still being generated when the loss of audio occurs. If you discover that the originating device stops sending packets for no reason, you might be dealing with a software or hardware problem on the originating device.

A common cause of such a failure is a Digital Signal Processor (DSP) crash. If the end device is a Cisco IOS software voice gateway, you see an error displayed on the console that looks similar to the following:

```
%VTSP-3-DSP_TIMEOUT: DSP timeout on event 6: DSP ID=0x2312: DSP error stats
```

This message is also sent to a Syslog server if the Cisco IOS software voice gateway is configured to send Syslog information to a Syslog server. On a Cisco VG200, 2600, or 3600, you can issue the following command to check the status of the DSPs:

```
test dsprm slot #
```

The **show voice dsp** command displays which port and time slot are allocated to each DSP. If the **test dsprm slot #** command detects a DSP that has crashed, you can compare this with the information obtained from a **show call active voice** command (or a **show call history voice** command if the call has been disconnected) to see if the time slot of the failed call is the same as the slot of the DSP that is no longer available. Unfortunately, the only way to recover from this condition is to reload the gateway.

Poor Voice Quality

Nearly all voice quality problems can be attributed to some kind of degradation on the IP network that the voice traffic traverses. Network problems that might not be noticeable for normal data traffic are very apparent in a voice conversation because of the need to minimize packet loss and variable delay in an IP telephony network.

A variety of issues can result in poor voice quality:

- [Packet Drops](#)
- [Queuing Problems](#)

In addition to the information in this section, refer to the [Troubleshooting QoS Choppy Voice Issues](#) document on Cisco.com for additional techniques on resolving voice quality issues.

Packet Drops

IP telephony demands that voice packets reach their destination within a predictable amount of time and without being dropped somewhere along the path from the source to the destination. In a properly designed network with appropriate QoS provisioning in place, packet loss should be near zero. All voice codecs can tolerate some degree of packet loss without dramatically affecting voice quality. Upon detecting a missing packet, the codec decoder on the receiving device makes a best guess as to what the waveform during the missing period of time should have been. Most codecs can tolerate up to five percent random packet loss without noticeable voice quality degradation. This assumes that the five percent of packets being lost are not being lost at the same time, but rather are randomly dropped in groups of one or two packets. Losing multiple simultaneous packets, even as a low percentage of total packets, can cause noticeable voice quality problems.



Note

You should design your network for zero packet loss for packets that are tagged as voice packets. A converged voice/data network should be engineered to ensure that only a specific number of calls are allowed over a limited-bandwidth link. You should guarantee the bandwidth for those calls by giving priority treatment to voice traffic over all other traffic. For more information on prioritizing voice over data, refer to the [Voice Quality](#) information available on Cisco.com.

There are various tools that you can use to determine whether you are experiencing packet loss in your network and where in the network the packets are getting dropped. The starting point to look for lost packets is the call statistics screen on Cisco Unified IP Phones.

1. Do one of the following:
 - If you are troubleshooting at the phone experiencing the problem, access these statistics by pressing the help (i or ?) button on the IP phone twice in quick succession during an active call.
 - If you are working with a remote user, open a web browser on your computer and enter the IP address of the user's phone. During an active call, choose the **Streaming Statistics > Stream 1** options from the display.
2. Examine the counters RxDisc and RxLost shown on the IP phone (or Rcvr Lost Packets if you are viewing the statistics remotely using a web browser).

- RxLost measures the number of packets that were never received because they were dropped in the network somewhere. By detecting a missing RTP sequence number, the IP phone can determine that a packet has been lost.
 - RxDisc corresponds to packets that were received but were discarded because they could not be used at the time they arrived. RxDisc can come from an out-of-order packet or a packet that arrived too late.
3. If either of these two counters increments, you should investigate to learn why packets are being lost or discarded.

Regardless of how low your packet loss is, if it is not zero, you should investigate the root cause because it might be a sign of a bigger problem that will get worse with higher call volume. Also, although small packet loss might not be perceptible in a conversation between two people, it can be detrimental to fax and modem transmissions. The packet loss can be occurring at any layer of the OSI model, so be sure to check for all possibilities for each hop. For example, if there is a Frame Relay connection over a T1 between two sites, you should:

- Make certain that there are no errors at the physical layer on the T1.
- Determine if you are exceeding your committed information rate (CIR) on the Frame Relay connection.
- Verify that you are not dropping the packets at the IP layer because you are exceeding your buffer sizes.
- Check that you have your QoS improperly configured.
- Ensure that your service provider not only guarantees packet delivery but also guarantees a low-jitter link. Some service providers may tell you that they do not provide a CIR but guarantee that they will not drop any packets. In a voice environment, delay is as important as packet loss. Many service providers' switches can buffer a large amount of data, thereby causing a large amount of jitter.

One common cause of drops in an Ethernet environment is a duplex mismatch, when one side of a connection is set to full duplex and the other side is set to half duplex. To determine if this is the case, perform the following steps:

1. Check all the switch ports through which a given call must travel and ensure that there are no alignment or frame check sequence (FCS) errors. Poor cabling or connectors can also contribute to such errors; however, duplex mismatches are a far more common cause of this kind of problem.
2. Examine each link between the two endpoints that are experiencing packet loss and verify that the speed and duplex settings match on either side.

Although duplex mismatches are responsible for a large number of packet loss problems, there are many other opportunities for packet loss in other places in the network as well. When voice traffic must traverse a WAN, there are several places to look. First, check each interface between the two endpoints, and look for packet loss. On all Cisco IOS software platforms, you can find this information using the **show interface** command. If you are seeing dropped packets on any interface, there is a good chance that you are oversubscribing the link. This could also be indicative of some other traffic that you are not expecting on your network. The best solution in this case is to take a sniffer trace to examine which traffic is congesting the link.

Sniffers are invaluable in troubleshooting voice quality problems. With a sniffer, you can examine each packet in an RTP stream to see if packets are really being lost and where in the network they are being lost. To troubleshoot using a sniffer, perform the following steps:

1. Start at the endpoint that is experiencing the poor-quality audio where you suspect packet loss.
2. Take a sniffer trace of a poor-quality call and filter it so that it shows you only packets from the far end to the endpoint that is hearing the problem. The packets should be equally spaced, and the sequence numbers should be consecutive with no gaps.

3. If you are seeing all the packets in the sniffer trace, continue taking traces after each hop until you get a trace where packets are missing.
4. When you have isolated the point in the network where the packet loss is occurring, look for any counters on that device that might indicate where the packets are being lost.

Queuing Problems

Queuing delay can be a significant contributor to variable delay (*jitter*). When you have too much jitter end-to-end, you encounter voice quality problems. A voice sample that is delayed over the size of the receiving device's jitter buffer is no better than a packet that is dropped in the network because the delay still causes a noticeable break in the audio stream. In fact, high jitter is actually worse than a small amount of packet loss because most codecs can compensate for small amounts of packet loss. The only way to compensate for high jitter is to make the jitter buffer larger, but as the jitter buffer gets larger, the voice stream is delayed longer in the jitter buffer. If the jitter buffer gets large enough such that the end-to-end delay is more than 200 ms, the two parties on the conference feel like the conversation is not interactive and start talking over each other.

Remember that every network device between the two endpoints involved in a call (switches, routers, firewalls, and so on) is a potential source of queuing or buffering delays. The ideal way to troubleshoot a problem in which the symptoms point to delayed or jittered packets is to use a sniffer trace at each network hop to see where the delay or jitter is being introduced.

For more information on jitter, refer to the [Understanding Jitter in Packet Voice Networks](#) document on Cisco.com.

PBX Interoperability Issues with Cisco Unified Communications Manager

If calls to destinations outside the IP network are failing, it may be because the calls must be routed through non-Cisco PBX switches that are connected to the PSTN. In order for the Cisco Unified Communications Manager to properly direct calls out trunks on the PBX, proper interoperability configuration is required. Information on configuring PBXs to interoperate with Cisco devices is available on an [Interoperability Portal](#) site. This site has information on third-party PBX interoperability with the following Cisco Collaboration products:

- Cisco Unified Communications Manager
- Cisco Unified Contact Center Enterprise
- Cisco Unified MeetingPlace

The information is provided in a series of application notes. If you cannot find your exact configuration in the list of available application notes, you may be able to use other application notes to meet your needs. Here are some suggestions:

- PBX product families should have similar results. For example:
 - PBXs in the Nortel Meridian 1 family should have similar results. The Nortel Meridian 1 Option 11C will have similar configurations as the Nortel Meridian 1 Option 61C and the Nortel Meridian 1 Option 81C.
 - PBXs in the Avaya Definity G3 family should be similar. The versions VSI, R, CSI, and SI should have similar configurations.
 - PBXs in the Siemens 300 family also should have similar configuration results. This family includes the 330, 340, and 370.

- Gateways configurations should be similar if the gateways are Cisco IOS based and have the same protocol (for example, MGCP, H.323, or SIP). For example, a Cisco 2801 gateway should have similar configurations to the 3845 when used with the Cisco Unified Communications Manager.

Using Call Flows to Resolve Call Processing Problems

This topic provides information about a typical call flow in an IP telephony environment. [Figure 6-1](#) shows a call flow that illustrates the actions in a typical call between the following devices at two different sites:

- Cisco Unified IP Phone (SCCP)
- Cisco Unified Communications Manager
- Gatekeeper

Figure 6-1 Call Flow in an IP Telephony Environment

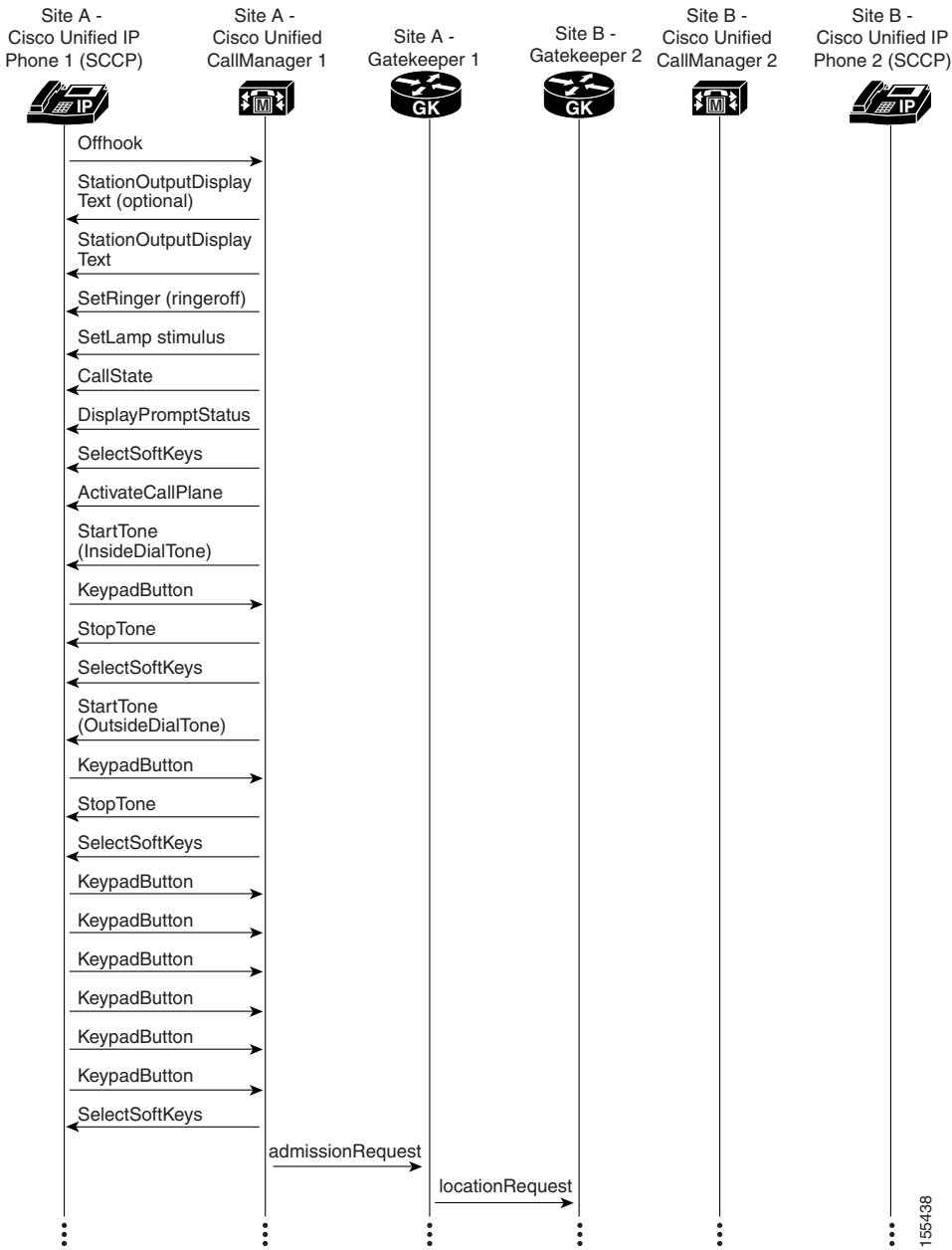
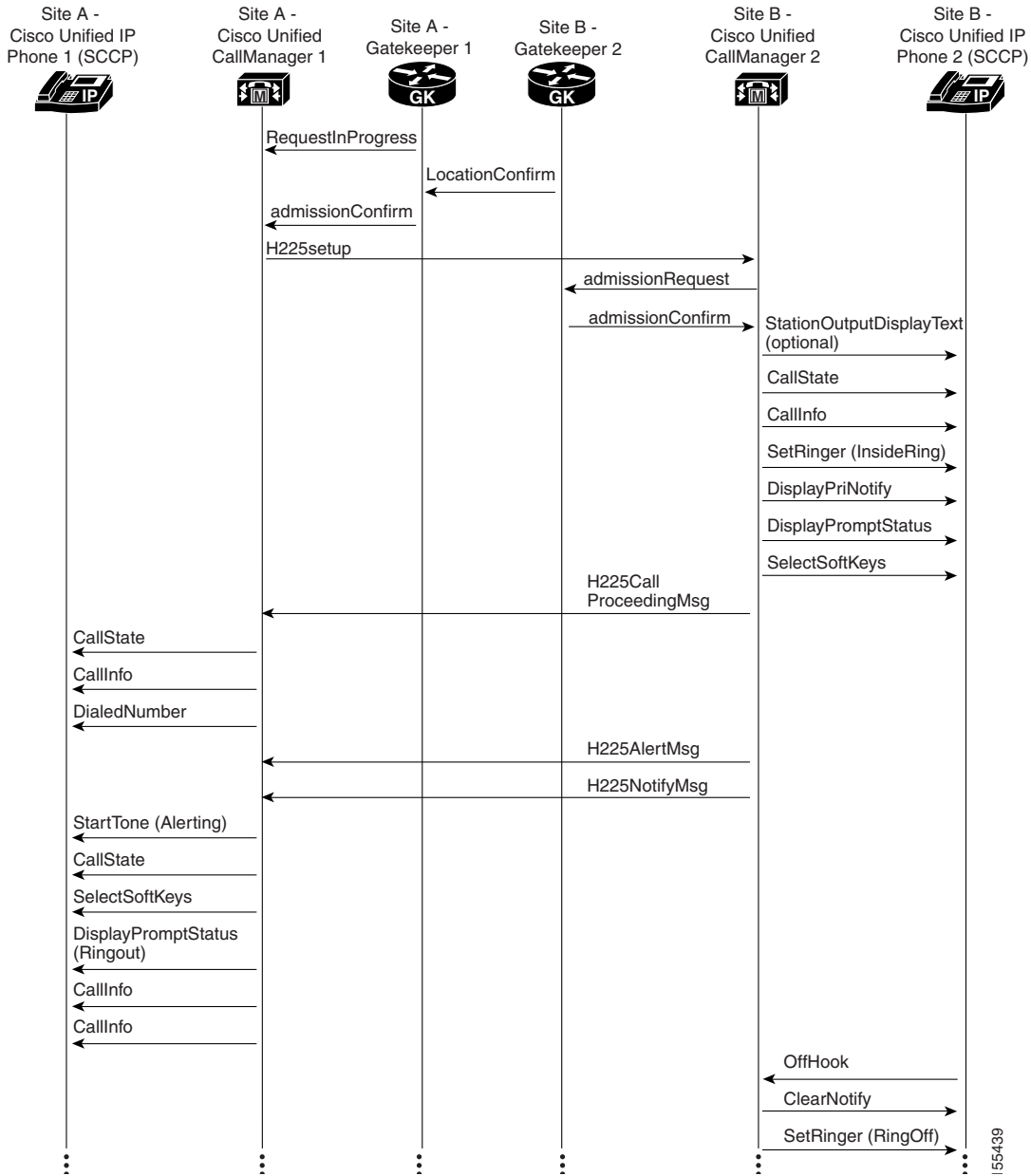


Figure 6-2 Call Flow in an IP Telephony Environment (continued)



155439

Figure 6-3 Call Flow in an IP Telephony Environment (continued)

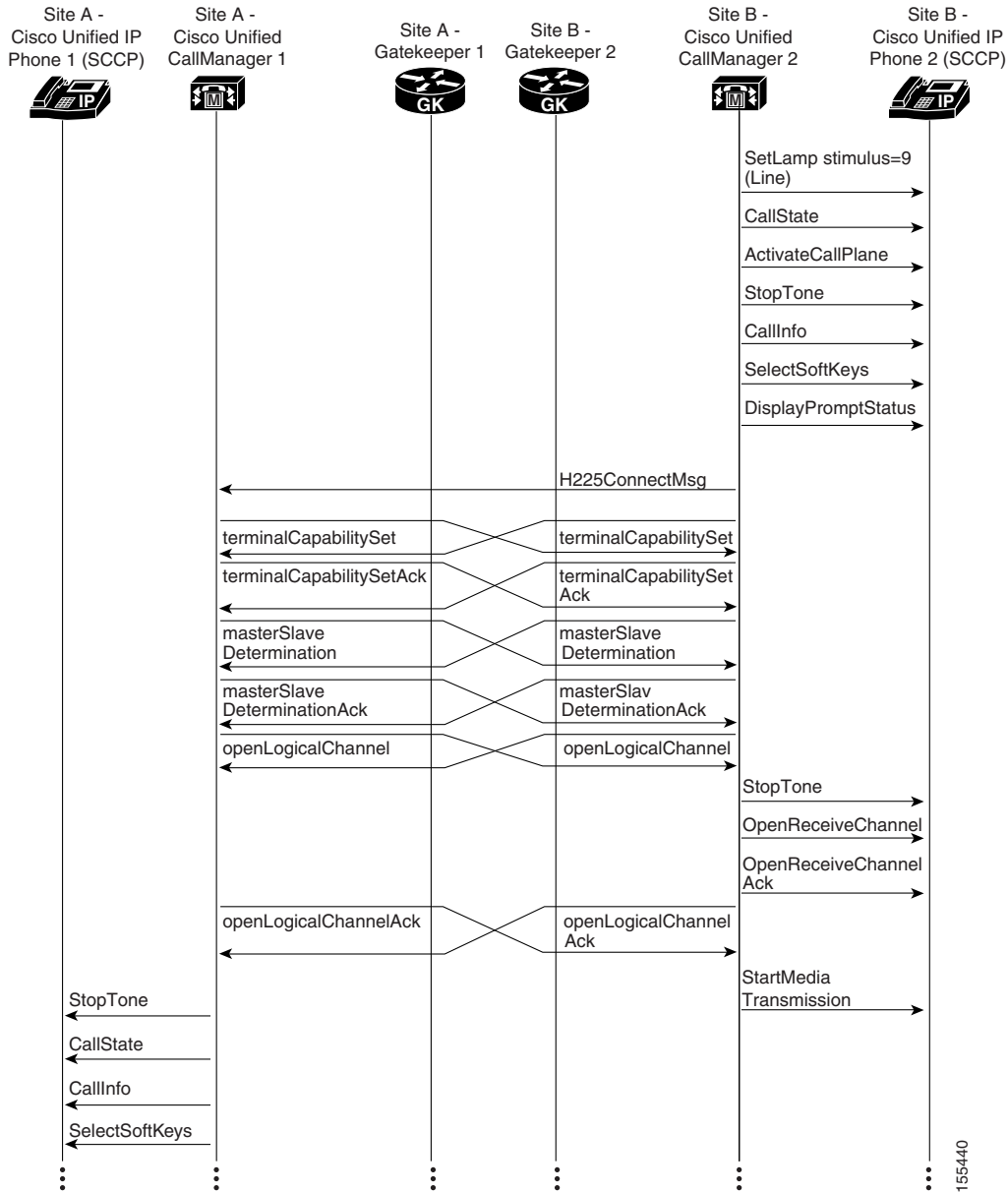


Figure 6-4 Call Flow in an IP Telephony Environment (continued)

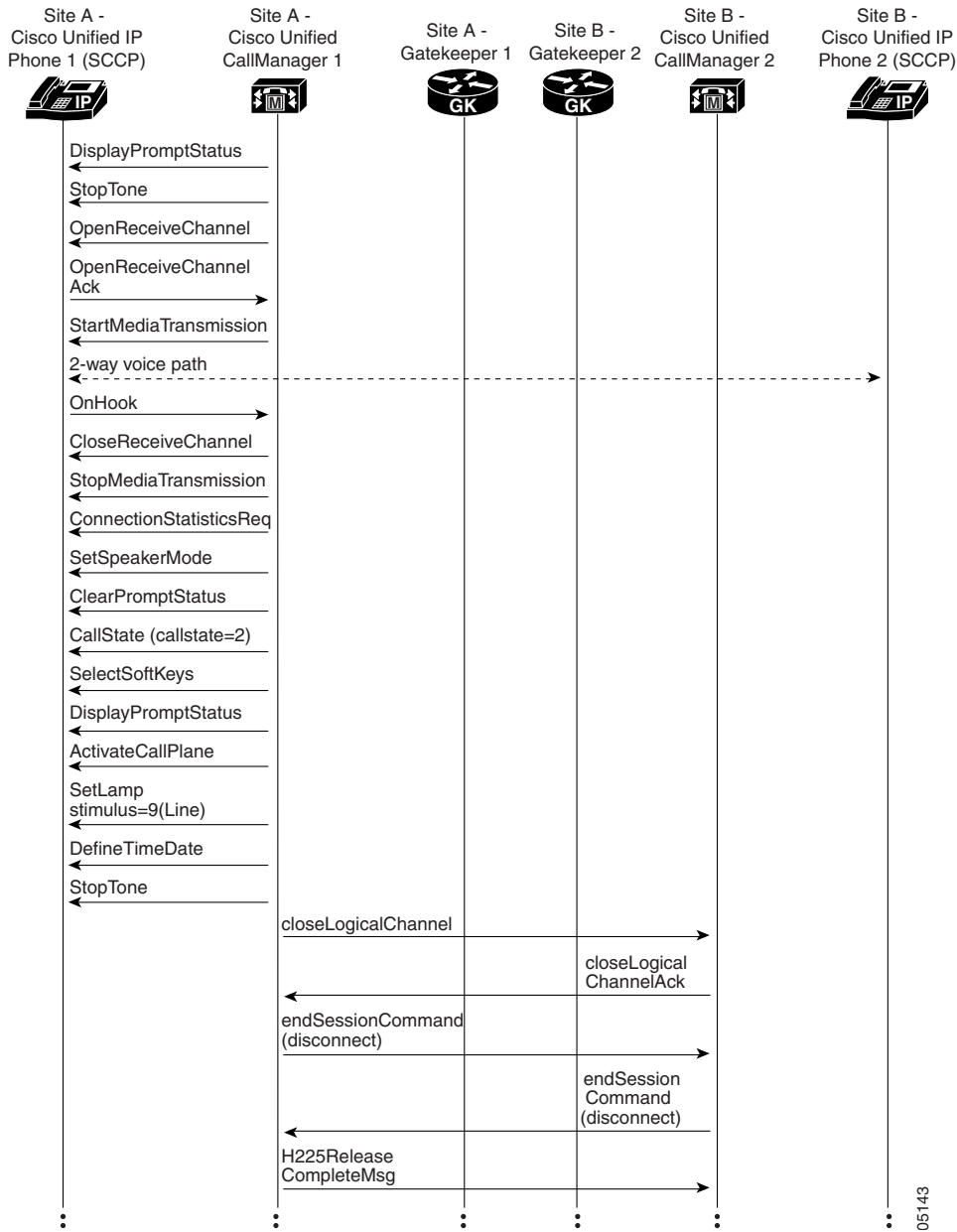
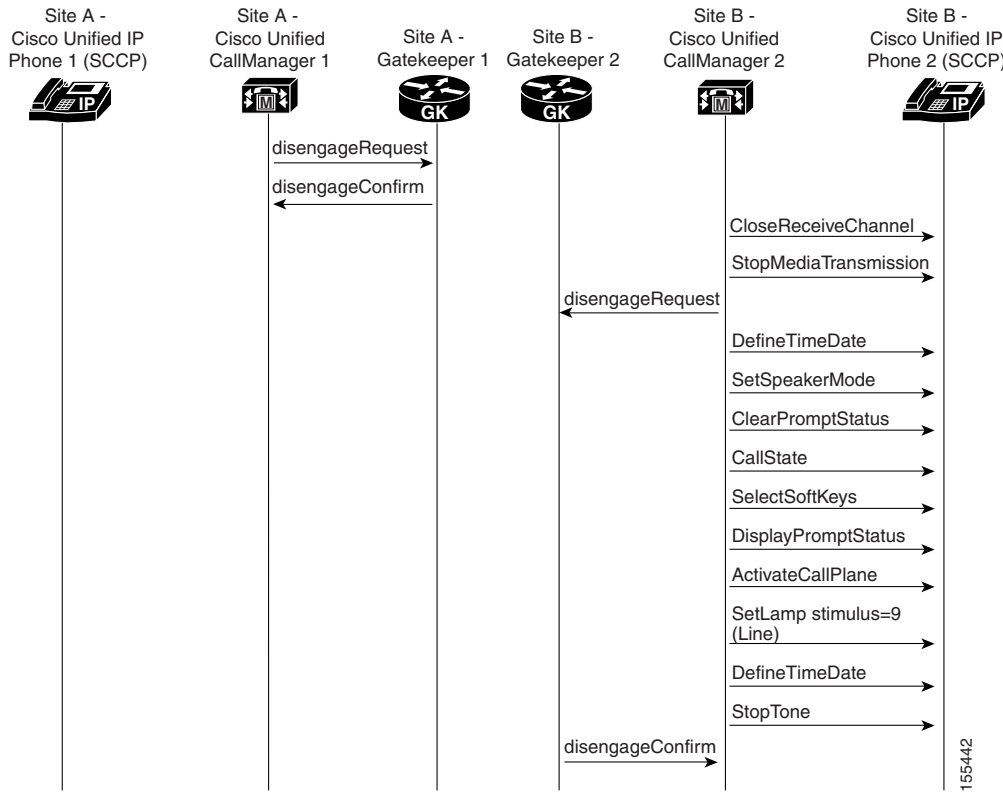


Figure 6-5 Call Flow in an IP Telephony Environment (continued)



For both successful and unsuccessful calls, an industry-standard cause code value appears in the disconnect or release signaling messages. The cause code reveals if the call was disconnected normally (typically cause code 16) or abnormally. Table 6-1 lists the standard cause values that may appear in the trace files as part of disconnect processing.

Table 6-1 Disconnect Cause Code Values

Hexadecimal Code with High-Order Bit Set	Hexadecimal Code Without High-Order Bit Set	Decimal Code	Description
0x81	0x01	1	Unallocated (unassigned) number. This cause indicates that the destination requested by the calling user cannot be reached because the number is unassigned. This number is not in the routing table, or it has no path across the ISDN network.
0x82	0x02	2	No route to specified transit network (National use). This number was dialed with a transit network code such as 108880 to go from AT&T and MCI, and there is no route across. The wrong transit network code was dialed.
0x83	0x03	3	No route to the destination. The dialed number is in the routing plan, but there is no physical route to the destination. The most likely cause of this is that the PRI D-channel is down, or the span or WAN is not connected correctly.
0x84	0x04	4	Send special information tone.

Table 6-1 Disconnect Cause Code Values (continued)

Hexadecimal Code with High-Order Bit Set	Hexadecimal Code Without High-Order Bit Set	Decimal Code	Description
0x85	0x05	5	Misdialed trunk prefix (National use).
0x86	0x06	6	Channel unacceptable.
0x87	0x07	7	Call awarded and being delivered in an established channel.
0x88	0x08	8	Preemption.
0x89	0x09	9	Preemption. Circuit reserved for reuse.
0x90	0x10	16	Normal call clearing. This is one of the most common codes and is received for many reasons. It usually occurs because someone hung up the call.
0x91	0x11	17	User busy. The number dialed is busy and cannot receive any more calls.
0x92	0x12	18	No user responding. The number that is being dialed has an active D-channel, but the far end chooses not to answer.
0x93	0x13	19	No answer from the user (user alerted). The number that is being dialed has an active D-channel, but the far end chooses not to answer.
0x94	0x14	20	Subscriber absent.
0x95	0x15	21	Call rejected.
0x96	0x16	22	Number changed. This cause code is generated when a subscriber on the PSTN has changed his or her phone number. Usually this message is accompanied by a progress indicator stating that in-band information is available. The PSTN provides an announcement in-band indicating the new phone number, if available.
0x9A	0x1A	26	Nonselected user clearing.
0x9B	0x1B	27	Destination is out of order. The number dialed is a working number, but the span is not active.
0x9C	0x1C	28	Invalid number format (address incomplete). This can happen when you are calling out using a network type number (enterprise) when you should be calling out Unknown or National.
0x9D	0x1D	29	Facility rejected.
0x9E	0x1E	30	Response to STATUS ENQUIRY.
0x9F	0x1F	31	Normal, unspecified. This is another common code. It happens when the network cannot determine what to do with the call being made.
0xA2	0x22	34	No circuit/channel is available. No B-channels are available to make the selected call.
0xA6	0x26	38	Network is out of order.
0xA7	0x27	39	Permanent frame mode connection is out of service.
0xA8	0x28	40	Permanent frame mode connection is operational.
0xA9	0x29	41	Temporary failure. The call was disconnected due to a network failure. This code appears for some long distance providers if the hunt sequence is incorrect. PRI lines must be set up for a flex hunt sequence (not a float hunt sequence).
0xAA	0x2A	42	Switching equipment congestion.

Table 6-1 *Disconnect Cause Code Values (continued)*

Hexadecimal Code with High-Order Bit Set	Hexadecimal Code Without High-Order Bit Set	Decimal Code	Description
0xAB	0x2B	43	Access information discarded. Usually reported when the far-end ISDN switch removes some piece of information before tandem-switching a call. For example, some PBXs strip the display IE before sending a call out toward the PSTN and send back a message with this cause code.
0xAC	0x2C	44	Requested circuit/channel is unavailable. This happens when you get in a glare condition: Both sides are selected top-down or bottom-up. Change the Allocation Direction (so that one end is top-down and the other is bottom-up).
0xAE	0x2E	46	Precedence call blocked.
0xAF	0x2F	47	Resource unavailable, unspecified. Whenever you see Cisco Unified Communications Manager initiate a disconnect with cause code 0xAF, 99% of the time the problem is related to a media setup failure. Check for codec capabilities mismatches, especially your regions configuration.
0xB1	0x31	49	Quality of service unavailable.
0xB2	0x32	50	Requested facility not subscribed. This code typically indicates you are trying to use a service you are not permitted to use. For example, you might be trying to make a voice call on an ISDN circuit provisioned for data only.
0xB5	0x35	53	Outgoing calls barred within Closed User Group (CUG).
0xB7	0x37	55	Incoming calls barred within CUG.
0xB9	0x39	57	Bearer capability not authorized. This code indicates that you are placing a call with a bearer capability you are not allowed to use.
0xBA	0x3A	58	Bearer capability not presently available. This code indicates that you are placing a call with a bearer capability for which the service provider does not currently have capacity to supply.
0xBE	0x3E	62	Inconsistency in designated outgoing access information and subscriber class.
0xBF	0x3F	63	Service or option unavailable, unspecified.
0xC1	0x41	65	Bearer capability not implemented. The cause could be one of the following occurrences: <ul style="list-style-type: none"> You need to change the PCM Type value to the setting appropriate for your country. This is the most common cause, especially in countries where G.711 A-law companding is the standard. If your gateway is configured for μ-law and the service provider or PBX is expecting A-law, you will see calls disconnected with this cause code. The central office (CO) does not understand an information element in the setup message. You are connected to a PBX and you are sending out a network type number when the switch accepts only Unknown or National. You are selecting European PRI and you have the progress indicators turned on when they should be off.
0xC2	0x42	66	Channel type not implemented.
0xC5	0x45	69	Requested facility not implemented.

Table 6-1 Disconnect Cause Code Values (continued)

Hexadecimal Code with High-Order Bit Set	Hexadecimal Code Without High-Order Bit Set	Decimal Code	Description
0xC6	0x46	70	Only restricted digital information bearer capability is available (National use).
0xCF	0x47	79	Service or option not implemented, unspecified.
0xD1	0x51	81	Invalid call reference value. This code indicates that the far-end switch did not recognize the call reference for a message sent by the gateway.
0xD2	0x52	82	Identified channel does not exist. This code indicates a call attempt on a channel that is not configured on the far end. This could happen if you are using a fractional PRI. As of Cisco Unified Communications Manager Release 3.3, fractional PRIs are no longer supported.
0xD3	0x53	83	A suspended call exists, but this call identity does not.
0xD4	0x54	84	Call identity in use.
0xD5	0x55	85	No call suspended.
0xD6	0x56	86	Call having the requested call identity has been cleared.
0xD7	0x57	87	User is not a member of CUG.
0xD8	0x58	88	Incompatible destination. The cause could be one of the following occurrences: <ul style="list-style-type: none"> • The number being dialed is not capable of the type of call. • You are calling a restricted line in unrestricted mode. • You are calling a POTS phone using unrestricted mode.
0xDA	0x5A	90	Nonexistent CUG.
0xDB	0x5B	91	Invalid transit network selection (National use).
0xDF	0x5F	95	Invalid message, unspecified.
0xE0	0x60	96	Mandatory information element is missing. The far-end switch states that a message was received missing an information element it considers to be mandatory per the Q.931 specification.
0xE1	0x61	97	Message type nonexistent or not implemented.
0xE2	0x62	98	Message is incompatible with the call state, or the message type is nonexistent or not implemented. This code is usually indicative of an ISDN protocol mismatch. Each ISDN protocol variant has a slightly different state machine based on the state machines defined in the Q.931 specification. If the two sides of an ISDN connection are not configured for the same protocol, one side might violate the other's call state machine. If an ISDN message is sent that is not expected in the current call state, this cause is generated.
0xE3	0x63	99	An information element or parameter does not exist or is not implemented.

Table 6-1 *Disconnect Cause Code Values (continued)*

Hexadecimal Code with High-Order Bit Set	Hexadecimal Code Without High-Order Bit Set	Decimal Code	Description
0xE4	0x64	100	Invalid information element contents. The cause could be one of the following occurrences: <ul style="list-style-type: none"> The call has an information element that is not understood by the switch being called. The E4 is usually followed by the information element that is causing the problem. The most common problem is that you are trying to place a call using a network number when the switch being called accepts only National, International, or Unknown dialing. This code is also generated when you are using Network-Specific Facilities as an element when they are not needed.
0xE5	0x65	101	The message is incompatible with the call state. This code is usually indicative of an ISDN protocol mismatch. Each ISDN protocol variant has a slightly different state machine based on the state machines defined in the Q.931 specification. If the two sides of an ISDN connection are not configured for the same protocol, one side might violate the other's call state machine. If an ISDN message is sent that is not expected in the current call state, this cause is generated.
0xE6	0x66	102	Recovery on timer expiry. This occurs when ISDN messages don't arrive in specified time according to the Q.931 specification. The E6 is sometimes followed by the timer that has expired (for example, 03 01 00—the 310 timer).
0xE7	0x67	103	Parameter nonexistent or not implemented—passed on (National use).
0xEE	0x6E	110	Message with unrecognized parameter discarded.
0xEF	0x6F	111	Protocol error, unspecified.
0xFF	0x7F	127	Interworking, unspecified.

Failover and Recovery Procedures

The Disaster Recovery System (DRS), which can be invoked from Cisco Unified Communications Manager Administration, provides full data backup and restore capabilities for all servers in a Cisco Unified Communications Manager cluster. The Disaster Recovery System allows you to perform regularly scheduled automatic or user-invoked data backups. DRS supports only one backup schedule at a time.

The Cisco Disaster Recovery System performs a cluster-level backup, which means that it collects backups for all servers in a Cisco Unified Communications Manager cluster to a central location and archives the backup data to physical storage device.

When performing a system data restoration, you can choose which nodes in the cluster you want to restore.

The Disaster Recovery System includes the following capabilities:


- A user interface for performing backup and restore tasks.
- A distributed system architecture for performing backup and restore functions.

- A scheduling engine to initiate tasks at user-specified times.
- Archive backups to a physical tape drive or remote sftp server.

The Disaster Recovery System contains two key functions, Master Agent (MA) and Local Agent (LA). The Master Agent coordinates backup and restore activity with all the Local Agents. The system automatically activates both the Master Agent and the Local Agent on all nodes in the cluster. However, you can only access the Master Agent functions on the first node of the cluster.

For more information on the Cisco Unified Communications Manager Disaster Recovery System, see the [Disaster Recovery System Administration Guide for Cisco Unified Communications Manager](#).

Additional Sites and Services

Steps to Success is a Cisco methodology that outlines the tasks required to complete a successful customer engagement. Registered users can visit the [Steps to Success](#)  resource site for Cisco Collaboration process flows.

Cisco Unified Communications Services is a Cisco service offering that provides engineering expertise and best practices.

- Registered users can visit the [Sell and Market Services](#)  partner site.
- Non-registered users can visit the [Services for Unified Communications](#) site.



CHAPTER 7

Optimize

Optimizing Your System

Optimization covers any changes to an existing system, including hardware and software upgrades, that enhance the functionality and performance of your network.

Collecting and analyzing data from your system's performance reports will provide crucial information for optimizing your system. By maintaining the routine system management procedures that you set up for your operations lifecycle, you will know when your traffic load increases, and when to increase your capacity.



Tip

You can navigate to any topic on this tab by using the tab navigation pane at the left of the content pane. This navigation pane contains the table of contents (TOC) for the active tab.

Input to This Process

Your network has been operational for some period of time and is ready to optimize based on system performance criteria. Your daily operations and growing business needs provide continuous feedback for optimization.

Output of This Process

User feedback, audits, and test results provide data to continue optimizing the system.

Major Tasks in This Process

- [Performing a System Upgrade](#)

Performing a System Upgrade


See [Preparing for Your System Upgrade](#) on the Plan tab to plan your overall strategy.

Upgrade IP Telephony Components

Once you have your upgrade plan and preparations in place, conduct your system upgrade following the guidelines and sequence in the individual product documentation (Refer to [Component Installation and Configuration Guides](#)):

- See the Collaboration Deployments Model chapter in [Cisco Collaboration Systems Solution Reference Network Designs \(SRND\)](#).
- See [Tested Deployments and Site Models](#) for a system level description of the tested deployment models.

Additional Sites and Services

Steps to Success is a Cisco methodology that outlines the tasks required to complete a successful customer engagement. Registered users can visit the [Steps to Success](#)  resource site for Cisco Unified Communications process flows.

Cisco Unified Communications Services is a Cisco service offering that provides engineering expertise and best practices.

- Registered users can visit the [Cisco Unified Communications Services](#)  partner site.
- Nonregistered users can visit the [Services for Unified Communications](#) site.



CHAPTER 8

Resource Library

Using the Resource Library

On this tab you find an assortment of resources to help you learn more about Cisco Collaboration Systems.

- [System Release Documentation](#)
- [Solution Reference Network Design](#)
- [Tested Deployment and Site Models](#)
- [Network Topology Diagrams](#)
- [Component Resources](#)
- [System Compatibility Tool](#)
- [Ordering Guides](#)
- [End-of-Sale and End-of-Life Products](#)
- [Cisco Unified Workspace Licensing](#)

System Release Documentation

The following system-level documents are produced for Cisco Collaboration Systems releases:

- [System Description](#)
- [System Release Notes](#)
- [System Release Summary Matrix](#)
- [Documentation Wiki](#)

System Description

This document provides an overview of the Cisco Collaboration Systems Release 10.0(1). It describes the Cisco Collaboration system-level approach, and lists key features of the Cisco Unified Communications components.

- [System Description for Cisco Collaboration Systems, Release 10.5\(1\)](#)

System Release Notes

These documents described the tested functionality, new and changed features, and limitations and restrictions for each of the following Cisco Collaboration Systems Release 10.0(1) solutions:

- Contact Center
 - [System Release Notes for Contact Center: Cisco Collaboration Systems, Release 10.5\(1\)](#)
- IP Telephony
 - [System Release Notes for Cisco Collaboration Systems, Release 10.5\(1\)](#)

System Release Summary Matrix

The summary matrix tables identify the software release versions of the system components that are used in each Cisco Collaboration Systems release for Contact Center and Collaboration.

[Cisco Collaboration Systems Release Summary Matrix for Contact Center](#)

[Cisco Collaboration Systems Release Summary Matrix for IP Telephony](#)

Documentation Wiki

The Cisco documentation wiki ([DocWiki](#)) contains information on a number of Cisco product-related topics. Among these is a category for Cisco Collaboration Systems, which currently includes topics for:

- [Unified Communications Systems Design](#): This topic includes information and tutorials on design tools such as the Cisco Collaboration Sizing Tool, which can be used to design and model solutions for existing and prospective customers.
- [Unified Communications Systems Implementation](#): This topic includes information on configuring system components, and provides detailed configuration examples based on tested deployment models.
- [Unified Communications Systems Operations](#): This topic includes information on the tasks you need to perform to maintain and optimize your system and keep it operating as trouble-free as possible. These tasks are broken down into two areas: one-time and infrequent tasks, and regular and scheduled tasks.
- [Unified Communications Systems Troubleshooting](#): This topic includes information that will assist you with isolating and resolving problems you may have with Cisco Collaboration Systems components. This topic offers sections for system troubleshooting methodology and commonly encountered problems.
- [Unified Communications Virtualization](#): This topic includes information on running Unified Communications applications on a virtual machine in a Unified Computing System (UCS) environment.

For additional Cisco Unified Communications topics posted on the Cisco documentation wiki, see http://docwiki.cisco.com/wiki/Cisco_Unified_Communications.

Solution Reference Network Design

Solution Reference Network Design (SRND) guides provide considerations and guidelines for deploying components for the Cisco Collaboration Systems. SRND resources are available at <http://www.cisco.com/go/srnd>.

Tested Deployment and Site Models

Cisco Collaboration Systems Release 10.5(1) testing was designed to test the hardware and software components that work together in a multisite distributed deployment. For this testing, several site models were created. Each site model was designed to test a specific set of features and interactions. For information about the components, configurations, and environment tested in this Cisco Unified Communications Release, see [Table 8-1](#).

Table 8-1 Tested Deployment and Site Models

Deployment Model	Description
Single-site Model	This model is designed for autonomous offices in which most or all employees are IPC users. This model can support up to 30,000 users.
Multisite centralized call processing model	This model is designed for distributed operations with a large central or headquarters site and multiple remote or branch sites. This model can support up to a total of 30,000 phones distributed among up to a maximum of 1000 sites. Based upon the bandwidth available, each site can support any number of users up to the overall total of 30,000 phones.
Multisite distributed call processing model	This model is designed for organizations with large user populations or large numbers of geographically distributed sites resulting in the need for more than a single call processing entity. This model is suited for deployments that require multiple Cisco Unified Communications Manager clusters or Cisco Unified Communications Manager Express platforms. Each call processing entity in this model is configured as a Single-Site Model or Multisite Centralized Call Processing Model and each has a common dial plan and feature set.
Clustering over IP WAN call processing model	This model is designed for organizations with large user populations across multiple sites that are connected by an IP WAN with the QoS features enabled. It supports the Local Failover Deployment Model and the Remote Failover Deployment Model.

For more information on the deployment models, refer to [Cisco Collaboration Systems 10.x Solution Reference Network Designs \(SRND\)](#).

Network Topology Diagrams

This topic provides topology resources that you can use to document network plans.

- For contact center solutions:

- [Network Topology Diagrams for Contact Center](#)
- For IP telephony solutions:
 - [Network Topology Diagrams for Collaboration](#)

You can download zip files containing Visio drawings of the logical and physical topologies.

Network Topology Diagrams for Contact Center

Table 8-2 *Network Topology Diagrams for Contact Center*

Description	Filename
Zip file includes logical and physical drawings in Test Bed as listed below.	CC_CSR1051_UCCE_Topology.zip (right-click to download zip) 4 MB Note If you click the link rather than right-click, the zip file may open in the secondary window but also replace the contents of this window. Use your Back button to return to this window.
Unified CCE components in Test Bed1	CC_CSR1051_UCCE_Topology.vsd
Zip file includes logical and physical drawings in Test Bed as listed below.	CC_CSR1051_UCCX_Topology.zip (right-click to download zip) 3 MB Note If you click the link rather than right-click, the zip file may open in the secondary window but also replace the contents of this window. Use your Back button to return to this window.
Unified CCX components in Test Bed2	CC_CSR1051_UCCX_Topology.vsd

Network Topology Diagrams for Collaboration

Table 8-3 *Network Topology Diagrams for Collaboration*

Description	Filename
Zip file includes the site model topology drawings listed below.	COL_CSR1051_Topology.zip (9) MB Note If you click the link rather than right-click, the zip file may open in the secondary window but also replace the contents of this window. Use your Back button to return to this window.
Collaboration Site in Test Bed	COL_CSR1051_Topology.vsd

Component Resources

For Cisco Collaboration Systems Release 10.0(1), there are two types of resources that provide descriptive information on each component in the solution:

- [Component Resources Documentation](#)
- [Configuration Command Files](#)

Component Resources Documentation

Depending on your Unified Communications solution type, choose one of the following sets of component resources documentation:

- [Component Resources Documentation for Contact Center](#)
- [Component Resources Documentation for Collaboration](#)

Component Resources Documentation for Contact Center

For links to the main page for documentation on Cisco Collaboration Systems Contact Center components, go to [Product documentation](#).

For documentation on specific tasks, see [Component Installation and Configuration Guides](#) on the Implement tab.

For documentation on all Cisco products, go to <http://www.cisco.com/cisco/web/psa/default.html?mode=prod>

Component Resources Documentation for Collaboration

For links to the main page for documentation on Cisco Collaboration Systems Contact Center components, go to [Product documentation](#).

For documentation on specific tasks, see [Component Installation and Configuration Guides](#) on the Implement tab.

For documentation on all other Cisco products, go to <http://www.cisco.com/cisco/web/psa/default.html?mode=prod>.

Configuration Command Files

To understand how various components were configured during Cisco Collaboration Systems testing, review the contents of the ZIP archives described in one of the following sections:

- [Configuration Command Files for Contact Center](#)
- [Configuration Command Files for Collaboration](#)

Additional system-level configuration examples are available on the Cisco DocWiki at: http://docwiki.cisco.com/wiki/Category:Unified_Communications_System_Implementation

Configuration Command Files for Contact Center

[Table 8-4](#) describes the contents of three ZIP files that you can download. Each ZIP file contains text files that contain the output from issuing a **show running-config** IOS command on various infrastructure components in the contact center test beds.

Table 8-4 Configuration Command Files for Contact Center

Description	Filename
Test Bed Local & Remote Agent Call Flows: Cisco Unified Customer Voice Portal (CVP) Post-Routed Call Flow Components	CC_CSR1051_UCCE_Configs.zip (right-click to download zip) includes the files below. Note If you click the link rather than right-click, the zip file may open in the secondary window but also replace the contents of this window. Use your Back button to return to this window.
Test Bed 2 Local & Remote Agent Call Flows: Cisco Unified Contact Center Express (Unified CCX) Call Flow Components.	CC_CSR1051_UCCX_Configs.zip (right-click to download zip) includes the files below. Note If you click the link rather than right-click, the zip file may open in the secondary window but also replace the contents of this window. Use your Back button to return to this window.

Configuration Command Files for Collaboration

[Table 8-5](#) describes the contents of a ZIP file with sample IP telephony configurations that you can download. Each ZIP file contains text files that contain the output from issuing a **show running-config** IOS command on various components in the IP telephony tested deployments.


Table 8-5 Configuration Command Files for Collaboration

Description	Filename
Zip file includes Cisco device configurations listed below: Note Some parameters, such as passwords, have been removed from the configuration files for security reasons.	COL_CSR1051_Configs.zip (right-click to download zip) includes the files below. Note If you click the link rather than right-click, the zip file may open in the secondary window but also replace the contents of this window. Use your Back button to return to this window.
Collaboration Site in Test Bed	lhr-srst-sip-01-config.txt sea-srst-sip-01-config.txt sfo-cube-asr-01-config.txt wdc-pstn-mgcp-01-config.txt

System Compatibility Tool

The [Cisco Collaboration Compatibility Tool](#) provides tables that identify the compatible software release versions for each product element in each Cisco Collaboration release.

Ordering Guides

[Ordering guides](#)  for most Cisco Collaboration products are available for Cisco partners, Cisco sales staff, and Cisco service providers.

End-of-Sale and End-of-Life Products

The end-of-sale (EOS) date is the last date to order the product through Cisco point-of-sale mechanisms. The product is no longer for sale. There is also an end-of-life (EOL) process that guides the final business operations associated with the product life cycle.

The EOL process consists of a series of technical and business milestones and activities that, once completed, make a product obsolete. Once obsolete, the product is not sold, manufactured, improved, repaired, maintained, or supported.

For information about recommended replacements, see the comprehensive list of announcements at the following URL:

http://www.cisco.com/en/US/products/prod_end_of_life.html

For information on specific products, choose a product from the following URL:

<http://www.cisco.com/en/US/netsol/ns1007/products.html>

Then click on the End-of-Life and End-of-Sale Notices link in the Product Literature section.

For an overview of the Products and Services EOL policy, see the information at the following URL:

http://www.cisco.com/en/US/products/products_end-of-life_policy.html

Cisco Unified Workspace Licensing

Cisco Unified Workspace Licensing is an easy, affordable program for procurement of a broad range of Cisco Unified Communications applications and services. Unified Workspace Licensing facilitates consistent deployment of multiple applications to all users in their workspaces and helps organizations maximize the potential of unified communications.

This program streamlines pricing, licensing, and deployment of Cisco Unified Communications solutions and enables organizations to implement a media-rich unified communications experience at a cost-effective user basis.





CHAPTER 9

Training Library

Using the Training Library

This tab includes training resources to help you learn more about Cisco Collaboration. The training resources are organized by audience:

- [General Training](#) contains links to courses and videos on demand (VoDs) for a general audience.
- [Training Available to Partners](#) contains resources for Cisco authorized partners and resellers.
- [Training Available to Cisco Employees](#) contains resources available for internal use only.

These categories list both online and instructor-led courses, downloadable VoDs, and links to online events and webcasts.


General Training

These Cisco training websites provide training on all Cisco products and technologies and are available to a general audience.

- [The Cisco Learning Network](#)
- [Global Learning Locator](#)
- [Featured Virtual Events and Webcasts](#)
- [Cisco Press self-study resources](#)

Training Available to Partners

Cisco offers a variety of training resources to partners. This topic describes some courses that are specific to Cisco Collaboration Systems Release 10.5(1).

Some Cisco Unified Communications training VoDs are available from the [Cisco Unified Communications System Release Technical Readiness](#)  page. Click the Tools & Training tab on the readiness page for more information on partner training.

For training information on all Cisco products and technologies, see the following sites.



Note

To access these sites, you must be registered as an employee of a Cisco Channel Partner company.

- [Partner Education Connection](#) 

Partner Education Connection Courses

Partner Education Connection (PEC) courses are the primary learning source for Cisco Channel Partners. PEC provides training on products, tools, and solutions.

Use the following e-learning modules for information on Cisco Collaboration Systems Release 10.5(1):


- QuickStart modules
These web-based modules prepare partners to sell Cisco Collaboration Systems solutions. This course covers selling strategies, pricing, and customer testimonials.
- Partner Beta Training
The PEC site typically contains a VoD which provides information on what is new in the latest Cisco Collaboration Systems.
- Cisco Collaboration Systems Architecture and Design
These web-based modules address identifying the features of the Cisco Collaboration Systems and teaching the Cisco methodology for implementing voice over data networks.

Cisco Learning Partner Courses

[Training from Cisco Learning Partners](#) provides a comprehensive set of training resources, from instructor-led courses to remote access labs and e-learning solutions. These companies are the only organizations to employ Certified Cisco Systems Instructors and deliver Cisco authorized and approved content, including product-specific training, technology training, and certification-preparation courses.

Training Available to Cisco Employees

Cisco offers a variety of training resources to employees. These Cisco Collaboration Systems launch page and training websites provide training on all Cisco products and technologies:

- [Partner Education Connection](#) 
- [Global Learning Partner Locator](#)



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