Cisco Training - HD Telepresence

CVOICE: Implementing Cisco Unified Communications Voice over IP and QOS 8.0

This 5-day course provides the learners with voice-related quality of service (QoS) mechanisms that are required in Cisco Unified Communications networks.

$3,495.00

- 5 Days
- Promotional and package discounts may apply

Upcoming Dates

Course Description

Implementing Cisco Voice Communications and QoS (CVOICE) teaches students about voice gateways, characteristics of VoIP call legs, dial plans and their implementation, basic implementation of IP phones in Cisco Unified Communications Manager Express environment, and essential information about gatekeepers and Cisco Unified Border Element.

The version of the authorized course materials is v8.0, but SLI is using 9.x in all our labs

Course Outline

Module 1: Introduction to Voice Gateway

Lesson 1: Understanding Cisco Unified Communications Networks and the Role of Gateways

- Cisco Unified Communications
- Cisco Unified Communications Gateways
- Gateways in Cisco Unified Communications Deployment Models
- Gateway Hardware Platforms
- Gateway Operational Modes

Lesson 2: Examining Gateway Call Routing and Call Legs

- Gateway Call-Routing Components
- End-to-End Call Routing
- Configuring POTS Dial Peers
- Dial Peer Matching
- Matching Inbound Dial Peers
- Matching Outbound Dial Peers
- Default Dial Peer
- Direct Inward Dialing

Lesson 3: Configuring Gateway Voice Ports

- Gateway Voice Ports
- Analog Voice Ports
- Configuring Analog Voice Ports
Lesson 4: Understanding DSP Functionality, Codecs, and Codec Complexity

- Voice Codecs
- Evaluating Quality of Codecs
- Evaluating Overhead
- Digital Signal Processors
- Codec Complexity
- Configuring DSPs
- Verifying DSPs

Module 2: VoIP Call Legs

Lesson 1: Examining VoIP Call Legs and VoIP Media Transmission

- VoIP Overview
- Converting Voice to VoIP
- VoIP Packetization
- VoIP Media Transmission
- Voice Activity Detection

Lesson 2: Explaining H.323 Signaling Protocol

- 323 Architecture
- 323 Call Flows
- Codecs in H.323
- Configuring H.323 Gateways
- Customizing H.323 Gateways
- Verifying H.323 Gateways

Lesson 3: Explaining SIP Signaling Protocol

- SIP Architecture
- SIP Call Flows
- SIP Addressing
- Codecs in SIP
- Configuring Basic SIP
- Configuring SIP ISDN Support
- Configuring SIP SRTP Support
- Customizing SIP Gateways
- Verifying SIP Gateways

Lesson 4: Explaining MGCP Signaling Protocol

- MGCP Architecture
- MGCP Call Flows
- MGCP Special Considerations
- Configuring MGCP Gateways
- Customizing MGCP Gateways
- Verifying MGCP Gateways

Lesson 5: Describing Requirements for VoIP Call Legs
Lesson 6: Configuring VoIP Call Legs

- Configuration Components of VoIP Dial Peer
- Configuring DTMF Relay
- Configuring Fax and Modem Support
- Configuring Codecs
- Limiting Concurrent Call

Module 3: Cisco Unified Communications Manager Express Endpoints Implementation

Lesson 1: Introducing Cisco Unified Communications Manager Express

- Introducing Cisco Unified Communications Manager Express
- Cisco Unified Communications Manager Express Key Features and Benefits
- Cisco Unified Communications Manager Express Supported Platforms
- Cisco Unified Communications Manager Express Operation

Lesson 2: Examining Cisco Unified Communications Manager Express Endpoint Requirements

- Overview of Cisco Unified Communications Manager Express Endpoints
- Identifying Cisco Unified Communications Manager Express Endpoint Requirements
- Power over Ethernet
- VLAN Infrastructure
- IP Addressing and DHCP Network Time Protocol
- Endpoint Firmware and Configuration
- Setting Up Cisco Unified Communications Manager Express in SCCP Environment
- Setting Up Cisco Unified Communications Manager Express in a SIP Environment

Lesson 3: Configuring Cisco Unified Communications Manager Express Endpoints

- Directory Numbers and Phones in Cisco Unified Communications Manager Express
- Creating Directory Numbers for SCCP Phones
- Configuring SCCP Phone-Type Templates
- Creating SCCP Phones
- Creating Directory Numbers for SIP Phones
- Creating SIP Phones
- Configuring Cisco IP Communicator Support
- Managing Cisco Unified Communications Manager Express Endpoints
- Verifying Cisco Unified Communications Manager Express Endpoints

Module 4: Dial Plan Implementation

Lesson 1: Introducing Call Routing

- Introducing Numbering Plans
- Scalable Numbering Plans
- Overlapping Numbering Plans
- Private and Public Numbering Plan Integration
- Number Plan Implementation Overview
- Call Routing Overview

Lesson 2: Understanding Dial Plans
Defining Dial Plans
Endpoint Addressing
Call Routing and Path Selection
PSTN Dial Plan Requirements
ISDN Dial Plan Requirements
Digit Manipulation
Calling Privileges
Call Coverage

Lesson 3: Describing Digit Manipulation
- Digit Collection and Consumption
- Components of Digit Manipulation
- Digit Stripping
- Digit Forwarding
- Digit Prefixing
- Number Expansion
- Caller ID Digits Manipulation
- Digit Manipulation Using Voice Translation Rules and Profiles
- Digit Manipulation Using dialplan-pattern Command
- Verifying Digit Manipulation

Lesson 4: Configuring Path Selection
- Call Routing and Path Selection
- Dial-Peer Matching
- Path Selection Strategies
- Site-Code Dialing and Toll Bypass
- Configuring Site-Code Dialing and Toll Bypass
- Tail-End Hop-Off
- Configuring TEHO

Lesson 5: Configuring Calling Privileges
- Calling Privileges Characteristics
- Implementing Calling Privileges on Gateways
- Implementing Calling Privileges on SRST and Cisco Unified Communications Manager Express
- Configuring COR
- Verifying COR

Module 5: Gatekeeper and Cisco Unified Border Element Implementation

Lesson 1: Understanding Gatekeepers
- Gatekeeper Overview
- Gatekeeper Signaling
- Gatekeeper Call Routing
- Gatekeeper-Based Call Admission Control
- Configuring Gatekeeper
- Configuring Gatekeeper Zones
- Configuring Zone Prefixes
- Configuring Technology Prefixes
- Adapting H.323 Gateways to Gatekeepers
- Configuring Gatekeeper CAC
- Verifying Basic Gatekeeper Functionality

Lesson 2: Examining Cisco Unified Border Element
- Cisco Unified Border Element Overview
Module 6: Quality of Service

Lesson 1: Introducing QoS

- QoS Issues
- QoS and Voice Traffic
- QoS for Unified Communications Networks
- QoS Requirements
- Methods for Implementing QoS Policy
- QoS Models

Lesson 2: Understanding QoS Mechanisms and Models

- DiffServ Model
- DSCP Encoding
- DiffServ PHBs
- DiffServ Class Selector
- DiffServ QoS Mechanisms
- Cisco QoS Baseline Model

Lesson 3: Explaining Classification, Marking, and Link Efficiency Mechanisms

- Modular QoS CLI
- Configuring Classification
- Configuring Class-Based Marking
- Trust Boundaries
- Mapping CoS to Network Layer QoS
- Link Efficiency Mechanisms Overview
- Link Speeds and QoS Implications
- Serialization Issues
- Link Fragmentation and Interleaving
- Configuring MLP with Interleaving
- Configuring FRF.12 Frame Relay Fragmentation
- Class-Based RTP Header Compression

Lesson 4: Managing Congestion and Rate Limiting

- Congestion and Its Solutions
- Policing and Shaping
- Measuring Traffic Rates
- Class-Based Policing
- Configuring Class-Based Policing
- Class-Based Shaping
- Configuring Class-Based Shaping
- Low-Latency Queuing
- Configuring LLQ
- Calculating Bandwidth for LLQ

Lesson 5: Understanding Cisco AutoQoS

- Cisco AutoQoS VoIP
- Configuring Cisco AutoQoS VoIP
- Monitoring Cisco AutoQoS VoIP
- Automation with Cisco AutoQoS
- Cisco AutoQoS for the Enterprise
- Configuring Cisco AutoQoS for the Enterprise
- Monitoring Cisco AutoQoS for the Enterprise

**Labs:**
- Lab 1-1: Configuring Voice Ports
- Lab 1-2: Configuring DSPs
- Lab 2-1: Configuring VoIP Call Legs
- Lab 3-1: Configuring Cisco Unified Communications Manager Express to Support Endpoints
- Lab 4-1: Implementing Digit Manipulation
- Lab 4-2: Implementing Path Selection
- Lab 4-3: Implementing Calling Privileges
- Lab 5-1: Implementing gatekeepers
- Lab 5-2: Implementing Cisco Unified Border Element
- Lab 6-1: Implementing QoS Using Cisco AutoQoS and Manual Configuration

**Audience**

This course is intended for:

- Network Administrators, Network Engineers, and Systems Engineers
- Anyone who has their CCNA Voice and/or is working towards CCNP Voice Certification

**Prerequisites**

To fully benefit from this course, students should have the following prerequisite skills and knowledge:

- Working knowledge of fundamental terms and concepts of computer networking, including LANs, WANs, and IP switching and routing.
- Ability to configure and operate Cisco IOS routers in an IP environment at the Cisco CCNA® Routing and Switching level.
- Basic knowledge of traditional voice, converged voice, and data networks at the Cisco CCNA Voice level

Recommended authorized prerequisite Cisco training: [ICND 1](#) & [ICND 2](#) or [CCNA Bootcamp](#), and [ICOMM](#)

**What You Will Learn**

After completing this course, students will be able to:

- Explain what a voice gateway is, how it works, and describe its usage, components, and features.
- Describe the characteristics and configuration elements of VoIP call legs.
- Describe how to implement IP phones using Cisco Unified Communications Manager Express.
- Describe the components of a dial plan, and explain how to implement a dial plan on a Cisco Unified voice gateway.
- Explain what gatekeepers and Cisco Unified Border Elements are, how they work, and what features they support.
- Describe why QoS is needed, what functions it performs, and how it can be implemented in a Cisco Unified Communications network.