

Cisco Unified Communications Manager Express 11.0

Collaboration Customized for the Branch Office

Cisco[®] Unified Communications Manager Express (Unified CME) provides call processing to Cisco Unified IP Phones for distributed enterprise branch-office environments and retail deployments. Even branch offices within the same enterprise can have different needs and requirements when it comes to unified communications. Cisco Unified Communications Manager Express delivers on this need by providing localized call control, mobility, and conferencing alongside data applications on Cisco Integrated Services Routers (ISRs).

Because the solution is Cisco IOS[®] Software-based, Cisco Unified Communications Manager Express is easy to configure and can be tailored to individual site needs. It is feature-rich and can be combined with Cisco Unity[®] Express and other services on the Cisco ISR to provide an all-in-one branch-office solution that saves valuable real estate space. Cisco Unified Communications Manager Express is ideal if you are looking for an integrated, reliable, feature-rich unified communications system for up to 450 users.

Key Features and Benefits

Unified communications is currently undergoing tremendous growth, accelerated by access to value-added features and applications only unified communications can provide. Additionally, the cost benefits of converging data, voice, and video onto a single network are adding to the rapid acceptance of this technology.

Cisco Unified Communications Manager Express enhances the advantages of convergence by offering the following benefits:

- Cost-effective operations through a single, integrated voice and data platform for all branch-office
 needs: Highly reliable Cisco routers, including the Cisco 800, 2900, 3900, and 4000 Series Integrated
 Services Routers, provide robust quality of service (QoS), network security, encryption, firewall, and
 network modules that deliver content networking and enhanced VPN services to address branch-office
 business needs. The system delivers integrated IP telephony, gateway, voicemail, and AutomatedAttendant functions, allowing you to deploy one platform to address all your business needs, thereby
 simplifying management, maintenance, and operations and delivering a lower total cost of ownership
 (TCO).
- Sophisticated key system and private-branch-exchange (PBX) capabilities: Retail offices have
 different workflows and require specialized features to support their work practices. Cisco Unified
 Communications Manager Express delivers a robust set of telephony features for the retail office and
 delivers innovative value-added multimedia capabilities through XML and Java midlets. These capabilities,
 which traditional systems cannot deliver, enhance the productivity of the end user and the business.

- Application integration: The Cisco Unified Communications Manager Express Services application
 programming interface (API) facilitates development of computer telephony integration (CTI) between Cisco
 Unified Communications Manager Express and third-party applications to enable call monitoring, call
 control, and call provisioning with any Cisco Unified CME Skinny Client Control Protocol (SCCP) endpoint.
 As a result, Cisco Unified CME can now be integrated with a variety of value-added unified communications
 service applications, in either co-located or service provider-hosted deployment scenarios, to support the
 critical unified communications services that are part of the end user's business process.
- Cisco Unified CallConnectors for desktop CTI: You can simplify communications and facilitate
 collaboration between users and customers with the easy-to-use interface of the Cisco Unified
 CallConnector suite.
- Contact-center capabilities: From basic call queuing to sophisticated contact centers for small to mediumsized companies, branch-office locations, or departments, you can deploy agent-assisted or self-service applications to reduce business costs and improve customer response by providing sophisticated and basic automatic call distributor (BACD), interactive voice response (IVR), CTI, and agent and desktop services.
- Interoperability with Cisco Unified Communications Manager: You can deploy Cisco Unified
 Communications Manager at larger sites and Cisco Unified Communications Manager Express at branchoffice locations where local call processing is required without a dependency on the WAN. Using H.323 or
 Session Initiation Protocol (SIP) trunking, you can route calls over the WAN with calling-party name and
 number information, plus compressed voice for better WAN bandwidth usage.
- Investment protection and ease of upgrade to centralized call-processing systems: With a simple software configuration change on the router, you can convert Cisco Unified Communications Manager Express to Cisco Unified Survivable Remote Site Telephony (SRST). Therefore, if you transition your unified communications architecture from a decentralized to a centralized architecture using Cisco Unified Communications Manager or Business Edition, the branch offices will be able to use Cisco Unified SRST to provide telephony survivability if a WAN failure occurs. Also, Cisco Unified SRST supports the Cisco Hosted Collaboration Solution (HCS) so that customers moving to cloud services can repurpose Cisco Unified CME for cloud telephony survivability. This flexibility helps ensure full investment protection no matter how a business's needs may change.
- Remote maintenance and troubleshooting: You can use the industry-standard Cisco IOS Software command-line interface (CLI) or user-friendly GUI to configure and administer Cisco Unified Communications Manager Express.

Cisco Unified Communications Manager Express allows a Cisco Integrated Services Router to provide rich call processing for Cisco IP phones including the Cisco Jabber[®] platform and the Cisco DX650 endpoints. All the necessary configurations and support files for IP phones are stored internally on the appliance, providing a single-platform solution. In addition, the solution offers a robust set of public-switched-telephone-network (PSTN) interfaces, integrated voicemail and Automated Attendant, and a full phone portfolio.

Cisco IOS Software offers industry-leading voice features designed for IP-based telephony systems, such as H.323 and SIP signaling, advanced quality of service (QoS), and ISR interworking with an H.323 gatekeeper or SIP proxy server—all available for use with Cisco Unified Communications Manager Express deployments. In addition, devices with integrated functions such as channel service unit/data service unit (CSU/DSU) and Network Termination 1 (NT1) are available with digital PSTN interface cards to provide flexible and robust voice services.

Cisco IP Phone Support

Using Cisco Unified Communications Manager Express, a maximum of 450 IP phones can be supported across a choice of platforms. IP phone operation is similar to that for Cisco Unified Communications Manager so that you can reuse the phones if you ever decide to migrate to a Cisco Unified Communications Manager, Business Edition, or Cisco HCS. Table 1 lists the maximum numbers of phones supported on each platform with Cisco Unified Communications Manager Express 11.0, and Table 2 lists the Cisco endpoints that Cisco Unified Communications Manager Express supports.

 Table 1.
 IP Phone Support per Platform

Platform	Maximum Number of Phones
Cisco 881 and 887VA Integrated Services Routers	5
Cisco 2901 Integrated Services Router	35
Cisco 2911 Integrated Services Router	50
Cisco 2921 Integrated Services Router	100
Cisco 2951 Integrated Services Router	150
Cisco 3925 Integrated Services Router	250
Cisco 3945 Integrated Services Router	350
Cisco 3925E Integrated Services Router	400
Cisco 3945E Integrated Services Router	450
Cisco 4321 Integrated Services Router	50
Cisco 4331 Integrated Services Router	100
Cisco 4351 Integrated Services Router	250
Cisco 4431 Integrated Services Router	350
Cisco 4451-X Integrated Services Router	450

Table 2. Cisco Endpoints Supported by Cisco Unified Communications Manager Express

IP Phone Series	Models Supported
Cisco IP Phone 7800 Series	 Cisco Unified IP Phone 7811 Cisco Unified IP Phone 7821 Cisco Unified IP Phone 7841 Cisco Unified IP Phone 7861
Cisco IP Phone 8800 Series	 Cisco IP Phone 8811 Cisco IP Phone 8821 Cisco IP Phone 8841 Cisco IP Phone 8851 Cisco IP Phone 8861 Cisco Unified IP Conference Phone 8831
Cisco Unified IP Phone 8900 Series	 Cisco Unified IP Phone 8961 Cisco Unified IP Phone 8941 Cisco Unified IP Phone 8945
Cisco Unified IP Phone 9900 Series	Cisco Unified IP Phone 9951Cisco Unified IP Phone 9971
Cisco Unified SIP Phone 3900 Series	Cisco Unified IP Phone 3905 Cisco Unified IP Phone 3911

IP Phone Series	Models Supported
Cisco Unified IP Phone 7900 Series	Cisco Unified IP Phone 7942G Cisco Unified IP Phone 7945G Cisco Unified IP Phone 7962G Cisco Unified IP Phone 7965G Cisco Unified IP Phone 7975G Cisco Unified IP Phone 7975G
	 Cisco Unified IP Phone Expansion Module 7915 and 7916 modules Cisco Unified IP Wireless IP Phone 7925G and 7926G models
Cisco DX Series	Cisco DX650
Cisco Jabber platform	Cisco Jabber for MAC and Windows

These intelligent Cisco Unified IP Phones support the following enhancements:

- · Display-based features with easy-to-use soft keys
- · Customer choice of using SCCP or SIP for commonly deployed IP phones
- · Language localization and feature customization along with support for XML-based applications
- Support for IEEE 802.3af or Cisco Prestandard Power over Ethernet (PoE) from a Cisco Catalyst[®] Switch, or Cisco EtherSwitch[®] Network Module or high-speed WAN interface card available on the Cisco 2900, 3900, and 4000 Series Integrated Services Routers

Product Features

Cisco Unified Communications Manager Express provides a sophisticated set of key system and PBX telephony features designed for businesses and branch offices. It also provides several industry-unique features that are not available with most other traditional telephony solutions. Table 3 summarizes the features available with Cisco Unified Communications Manager Express Version 11.0. Consult the Feature Navigator at http://www.cisco.com for the latest Cisco IOS Software release.

 Table 3.
 Cisco Unified Communications Manager Express 11.0 Features

Feature	Description
Phone features	Maximum 450 phones per system
	Up to 34 line appearances per phone
	Attendant console functions using Cisco Unified IP Phone Expansion Module 7915 and 7916 modules
	Fast transfer: Blind or consult
	Busy Lamp Field (BLF)
	Silent ringing options
	Distinctive ring per line
	Automatic line selection for outbound calls
	Call forward on busy, no answer, and all (internal or external)
	Call-forward-all restriction control
	Do not disturb (DND)
	Feature ring with DND set
	IP phone display of DND state
	Dial-plan pattern load on SIP phones
	Diversion of calls directly to voicemail
	Customization of soft keys
	Enable and disable call-waiting notification per line
	Call waiting with overlay directory number
	Call-waiting ring
	Dual or eight call line appearances per button
	After-hours toll-bar override
	Auto answer with headset

Feature	Description
	European date formats
	Hook flash pass-through across analog PSTN trunks
	Idle URL: Periodically push messages or graphics on IP phones
	Last-number redial
	Live record to Cisco Unity Express mailbox
	Local name directory lookup
	On-hook dialing
	Station speed dial with configuration changes from IP phone
	System speed dial for 10,000 numbers
	Silent and feature ring options
	SIP-based line-side subscribe, providing basic presence of phone status
	Transfer to voicemail soft key
	Call barge with privacy on shared lines
	Access features using soft keys or feature access codes
	Remote teleworker IP phone support
	Dynamic hunt-group join or leave
	Support for analog phones using Cisco ATA 186 Analog Telephone Adaptor or Cisco VG224 Analog Voice Gateway in SCCP mode
	Support for fax machines on foreign-exchange-station (FXS) ports or ATA using H.323, SCCP, or SIP
	Support for Cisco VG202 and VG204 Analog Voice Gateways
	XML application services on Cisco Unified IP display phones
	Station-to-station video with voice using Cisco Unified Video Advantage or Cisco Unified IP Phone 7985G endpoints
	Extension mobility within the single site
	Wideband audio (G.722) and Internet low bit rate codec (iLBC)
	Single number reach for mobility in Cisco IOS Software
	Whisper intercom
	Shared line on SIP phones with privacy and barge-in capabilities
	Customizable phone telephony user interface (TUI) with button layout control
	Improved security with customizable services and directories page
	Programmable line keys (PLK) on Cisco IP Phone TUI
	My Phone Apps support for BLF speed-dial configuration
	Normalized +E.164 support for Cisco Unified IP Phones (SCCP)
	Support for Cisco Mobile 8.1 for iPhone and iPod touch soft-phone client
	Secure Sockets Layer (SSL) VPN client support for Cisco IOS Datagram Transport Layer Security (DTLS) VPN for SCCP phones
	Localization support for Cisco Unified IP Phone 7920 and 6900 models with SCCP
	 Video and camera support for Cisco Unified IP Phone 8961, 9951, and 9971 models
	Extension Mobility support for SIP endpoints
	Support for Cisco Unified IP Phone 8941 and 8945 endpoints
	Support for Single Number Reach and paging for SIP endpoints
	Support for shared line between SCCP and SIP endpoints
	My Phone App support for SIP endpoints
	Support for Cisco ATA 187, ATA 188 and ATA 190 Analog Telephone Adapters
	Support for SIP-initiated conference
	Support for visual list of parked calls
	Support for visual list of voice hunt groups
	Support for dynamic hunt-group login
	Support for distinctive ring for parked call recall
	Support for audible ringtone with hunt-group login and logout
Trunk features	Analog foreign-exchange-office (FXO) Loop and Ground Start For and mouth (FSN)
	Ear and mouth (E&M) Pagis Pata Interface (PRI) and Primary Pata Interface (PRI) curport (NI2 4ESS EESS EuraISDN DMS100 and
	 Basic Rate Interface (BRI) and Primary Rate Interface (PRI) support (NIZ, 4ESS, 5ESS, EuroISDN, DMS100, and DMS250) and several other switch types currently supported in Cisco IOS Software
	Caller ID name and number Automatic group has identification (ANII)
	Automatic number identification (ANI) CT/(Tx)
	Digital trunk support (T1/E1)

Feature	Description
	Direct inward dialing (DID)
	Direct outward dialing (DOD)
	• E1 R2 support
	Dedicated trunk mapping to phone button
	H.323 trunks with H.450 support
	H450.12 automatic detection of H.450 support for remote H.323 endpoints
	H.323-to-H.323 hairpin call routing for non-H.450-compliant H.323 endpoints
	SIP trunks and RFC 2833 support
	• Transcoding with G.711, G.729a, and iLBC
	Call forwarding busy, no answer, and all
	Calling line identification presentation (CLIP) and calling name identification presentation (CNIP)
	Connected line identification presentation (COLP) and connected name identification presentation (CONP)
	Message-waiting indicator (MWI) and message center support
	MWI pass-through QSIG-to-time-division multiplexing (TDM) voicemail
	Overlap sending support on ISDN PRI and BRI trunks
System features	Account codes and call-detail-record (CDR) field entry
	Call-back busy subscriber and camp-on
	Per-phone call-coverage rules
	Call hold and retrieve
	Call park: Personal and directed
	Call transfer and park recall
	Call park assign to extension
	Call pickup directed
	Call pickup local group
	Call pickup explicit group
	Call transfer: Consultative and blind
	Call waiting
	Cancel call waiting soft key
	Call park and pickup on both SCCP and SIP phones
	CTI with Microsoft CRM and Outlook using Cisco IOS Software Telephony Services Provider (TSP)
	E911 with two emergency location numbers per zone; unlimited zones per site
	Eight-party impromptu conferencing
	Directory services using XML
	Hunt groups: Sequential, circular, parallel (blast), and longest idle
	Hunt-group dynamic log in and log out
	Hunt-groups statistics: Daily and hourly
	• Intercom
	Ad-hoc conferencing (8 parties)
	Meet-me conferencing (32 parties)
	• Five music-on-hold (MoH) streams (internal)
	Night service bell or call forwarding
	Overlay extensions for enhanced call coverage
	Called-name display for overlay extensions
	Paging: Internal through IP phones or to external system
	Per-call caller ID blocking
	Secondary dial tone
	Standards-based network call transfer and call forwarding using H.450
	Additional system speed-dial option through XML service
	Time-of-day and day-of-week call blocking
	Customizable called-name display
	Support of SRST fallback service phone autoregistration
	Basic automatic call distributor (BACD) (three queues) with Automated Attendant and call statistics
	Display of number of calls in queue on IP phone
	Agent log in and log out of BACD hunt group
	• Integration with Cisco Unified Contact Center Express 5.0 for advanced call-center features with support for up to

Feature	Description
	Secure Real-Time Transport Protocol (SRTP) providing media encryption for calls on the IP network
	Secure voice IP phone certificate authentication and provisioning plus secure device signaling using Transport Layer Security (TLS)
	Video over SIP trunk with H.264 codec support
	Multilevel Precedence and Preemption (MLPP) support over PRI trunks
	Enhanced MLPP with support for supplementary services, including three-way conference, call park, and call pickup
	BLF monitoring of conference directory numbers and call-park slots
	Cisco Unified Communications Manager Express and Cisco Unity Express username and password synchronization
	Support for IPv6 on SCCP and SIP IP phones
	Enhanced security with Logical Partition Class of Restriction for Closed User Group (CUG)
	Optional unlock of meet-me conference bridge
	New IP Trusted Authentication List for enhanced security
	Forced Authorization Code (FAC) for improved call screening and accounting
	SSL VPN support for Cisco Unified IP Phones
	SSL VPN support for Cisco Adaptive Security Appliance (ASA)
	Fixed mobile convergence (FMC) with media flow around on SIP trunks
	Support for Call Forward Unregistered
	Clear directory entries for calls
	Localization support for Cisco Unified IP Phone 7920 and 6900 models with SCCP
	 Video and camera support for Cisco Unified IP Phone 8961, 9951, and 9971 models
	Increased translation rule from 15 to 100
	Bulk registration support for SIP phones
	Debug ephone messages enhancement
	Support for Cisco Unified Wireless IP Phone 7926 wireless phone
	Increased fast dial limit to 100
	Support for "show voice register pool type summary"
	Support for conference maximum length
	Support for DX650 desktop collaboration endpoint
Voicemail features	Integrated voicemail and Automated-Attendant solution with Cisco Unity Express
	 Integration with Cisco Unity Voicemail and Cisco Unity unified messaging, or third-party voicemail integration (H.323, SIP, or dual-tone multifrequency [DTMF])
International localization	 Per-phone localization for up to five local languages per system, including English, Bulgarian, Chinese Mandarin and Cantonese, Croatian, Czech, Danish, Dutch, European Spanish, Finnish, French, German, Greek, Hungarian, Italian, Japanese Kanji and Katakana, Korean, Norwegian, Polish, Portuguese, Romanian, Russian, Serbian, Slovakian, Slovenian, Swedish, Turkish, Latvian, Lithuanian, Estonian, and Hebrew
Management features	Automatic assignment of extensions to phones for easy phone additions
	Extension assigner, allowing for deployment of new phones using voice prompts
	Single web-based GUI for moves, adds, and changes for system and integrated voicemail with three levels of GUI administration: System administrator, customer administrator, and user
	Centralized network management deployments using Cisco CNS Configuration Engine
	Telephony-service setup and configuration using HTML Quick Configuration Tool
	 Simple Network Management Protocol (SNMP) support with Cisco Unified Operations Manager or third-party management consoles
APIs for development and integration	 Client integration: You can connect Cisco Unified CME with Microsoft CRM and Salesforce.com using Telephony Application Programming Interface (TAPI) Windows driver based on Cisco IOS Software TSP.
	Server integration: You can connect Cisco Unified CME with third-party applications using the Cisco Unified Communications Express Services Interface API. This API enables protocol interaction between Cisco Unified CME and third-party applications to perform CTI functions, including call monitoring, call control, and call provisioning with any SCCP device registered with the Cisco Unified CME. It includes two elements: embedded CTI protocol support within Cisco Unified CME and Java Function Library (referred to as the "UC Express Services SDK") that co-resides on an external server with the third-party application. This API can support either co-located or service provider-hosted application deployment scenarios. For more information, please refer to: http://developer.cisco.com/web/ucxapi/home .

Summary

Cisco Unified Communications Manager Express delivers telephony features you need to meet the requirements of your branch office. Cisco Integrated Services Routers offer high-reliability and advanced applications, including unified communications, VPN, firewall, encryption, dial access, Ethernet switching with Power over Ethernet, and content networking within a single all-in-one platform that is easy to deploy and maintain, resulting in a lower TCO.

As your business expands, you can easily migrate Cisco Unified Communications Manager Express to a Cisco Unified Communications Manager larger-scale IP telephony solution. All hardware and software used by this solution is fully compatible with Cisco Unified Communications Manager and Cisco Unified SRST, giving you robust investment protection.

Cisco Unified Communications Services

Cisco Unified Communications Services allows you to accelerate cost savings and productivity gains associated with deploying a secure, resilient Cisco Unified Communications Solution. Delivered by Cisco and our certified partners, our portfolio of services is based on proven methodologies for unifying voice, video, data, and mobile applications on fixed and mobile networks. Our unique lifecycle approach to services enhances your technology experience to accelerate true business advantage.

Cisco Capital

Financing to Help You Achieve Your Objectives

Cisco Capital can help you acquire the technology you need to achieve your objectives and stay competitive. We can help you reduce CapEx. Accelerate your growth. Optimize your investment dollars and ROI. Cisco Capital financing gives you flexibility in acquiring hardware, software, services, and complementary third-party equipment. And there's just one predictable payment. Cisco Capital is available in more than 100 countries. Learn more.

For More Information

For more information about Cisco Unified Communications Manager Express, visit http://www.cisco.com/go/cme.

Partners looking for part numbers and ordering information should refer to the Cisco Unified Communications Manager Express ordering guide available under the sales resource <u>listing</u>.



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