



# Cox Business SIP Trunk: Connecting Cisco Unified Communication Manager Express (CME) v12.6 [IOS-XE 16.12.01a] using SIP

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## Introduction

Service Providers today, such as Cox Business, are offering alternative methods to connect to the PSTN via their IP network. Most of these services utilize SIP as the primary signaling method and a centralized IP to TDM gateway to provide on-net and off-net services.

Cox is a service provider offering that allows connection to the PSTN and may offer the end customer a viable alternative to traditional PSTN connectivity via either analog or T1 lines. A demarcation device between these services and customer owned services is recommended.

- This application note describes how to configure a Cisco Unified Communications Manager Express (Cisco Unified CME) v12.6, Cisco Unity Connection 12.0.1 with connectivity to Cox SIP trunk service. The application note also covers support and configuration example of Cisco Unity Connection (CUC) messaging integrated into the Cisco Unified CME. The deployment model covered in this application note is Customer Premises Equipment (Cisco Unified CME/CUC) to PSTN (Cox). Cox provides inbound and outbound call service.
- Testing was performed in accordance to Cox test plan and all features were verified. Key features verified are: inbound and outbound basic call (including international calls), calling name delivery, calling number and name restriction, CODEC negotiation, intra-site transfers, intra-site conferencing, call hold and resume, call forward (forward all, busy and no answer), leaving and retrieving voicemail (Cisco Unity Connection), Cisco auto-attendant (Cisco Unity Connection), fax using T.38 and G.711.
- Consult your Cisco representative for the correct IOS-XE image and for the specific application and Device Unit License and Feature License requirements for all your Cisco Unified CME.



## Network Topology

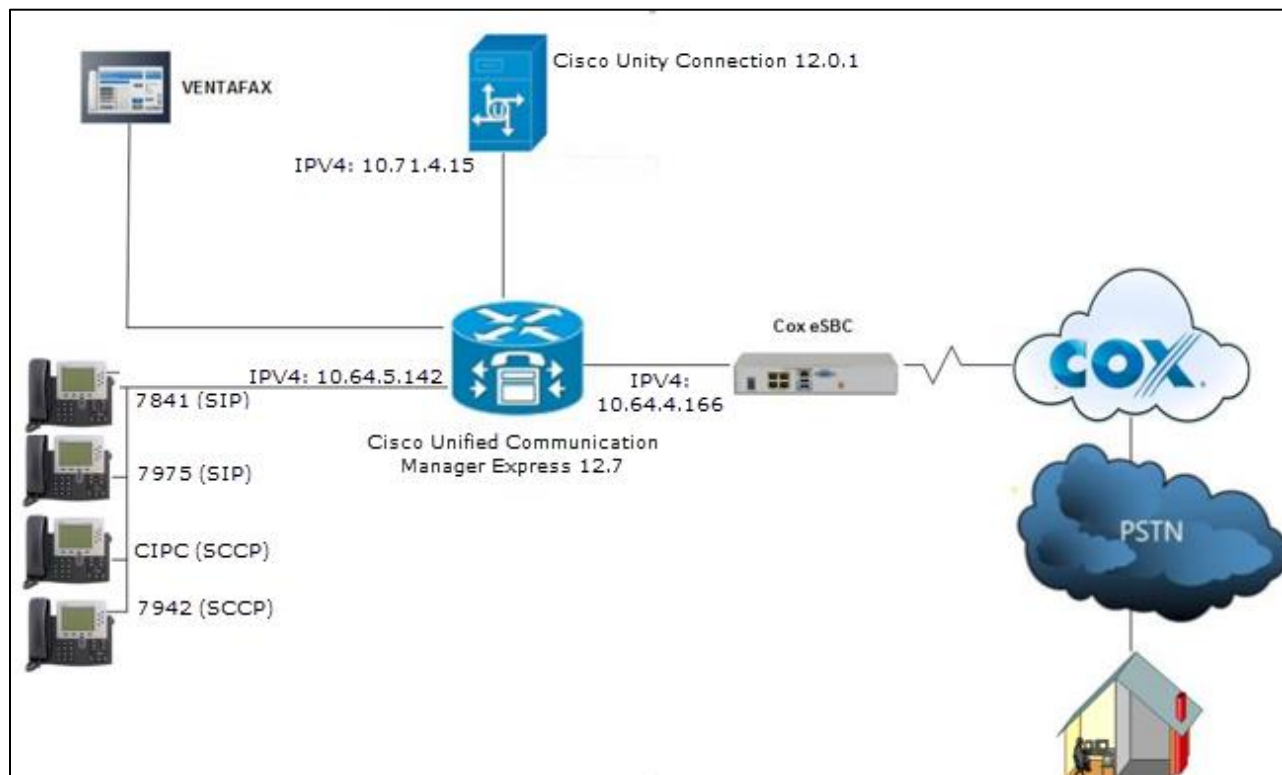


Figure 1 Network Topology

- The network topology includes the Cisco Unified CME, Unity Voicemail system and 4 Cisco Endpoints. Cisco Unified CME has a trunk configured to Cox IP Address. Cox was used as the service provider with SIP trunk to the Cox eSBC LAN IP Address.
- SIP Trunk transport type used between Cisco Unified CME to Cox eSBC is UDP.

### Cisco Unified CME Settings:

Setting	Value
Transport from Cisco Unified CME to Cox eSBC	UDP with RTP
Voice Mail Support	YES
Session Refresh	YES



## System Components

### Hardware Requirements

- Cisco ISR4321
- Cisco IP Phones: 7841, 7975, CIPC and 7942 phones, but any Cisco IP Phone model supporting RFC2833 can be used
- Cisco ISR4321/K9 (1RU) processor with 3743802K/3071K bytes of memory. Processor board ID FLM2201V1AX
- 2 Gigabit Ethernet interfaces and 2 Voice FXS interfaces
- Cisco Unity Connection - VMware - 1 vCPU: Intel(R) Xeon(R) CPU X5675 @ 3.07GHz
- HDD 160 GB, Memory 4096 Mbytes RAM

### Software Requirements

- Cisco IOS gateway running Cisco Unified CME 12.6 Version 16.12.01a, RELEASE SOFTWARE (fc2).  
Cisco IOS image: "bootflash:isr4300-universalk9.16.12.01a.SPA.bin"
- Cisco Unity Connection version 12.0.1.22900-14

## Features

### Features Supported

- Incoming and Outgoing off-net calls using G711ulaw
- Call Conference
- Voice Mail support
- Call Hold & Resume with and without MOH
- Unattended and Attended Call Transfer
- Call Forward (all, busy and no answer)
- DTMF (RFC2833)
- Fax (T.38 and g711ulaw pass-through)
- Calling Party Number Presentation and Restriction
- Auto Attendant

### Features Not Supported

- G.729 codec
- Blind Call Transfer
- SG3 speed for Fax



## Caveats

- None



## Configuration

### Cisco Unified Communication Manager Express

#### Cisco IOS Version

```
COXCME#sh version
Cisco IOS XE Software, Version 16.12.01a
Cisco IOS Software [Gibraltar], ISR Software (X86_64_LINUX_IOSD-UNIVERSALK9-M),
Version 16.12.1a, RELEASE SOFTWARE (fc2)
Technical Support: http://www.cisco.com/techsupport
Copyright (c) 1986-2019 by Cisco Systems, Inc.
Compiled Sun 04-Aug-19 06:12 by mcpre
```

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ROM: IOS-XE ROMMON

```
COXCME uptime is 2 days, 11 hours, 49 minutes
Uptime for this control processor is 2 days, 11 hours, 51 minutes
System returned to ROM by PowerOn at 06:27:44 UTC Thu May 14 2020
System restarted at 20:55:13 UTC Mon May 18 2020
System image file is "bootflash:isr4300-universalk9.16.12.01a.SPA.bin"
Last reload reason: PowerOn
```

This product contains cryptographic features and is subject to United States and local country laws governing import, export, transfer and use. Delivery of Cisco cryptographic products does not imply third-party authority to import, export, distribute or use encryption. Importers, exporters, distributors and users are responsible for compliance with U.S. and local country laws. By using this product you agree to comply with applicable laws and regulations. If you are unable to comply with U.S. and local laws, return this product immediately.

A summary of U.S. laws governing Cisco cryptographic products may be found at:  
<http://www.cisco.com/wwl/export/crypto/tool/stqrg.html>

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

Suite License Information for Module:'esg'

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Suite	Suite Current	Type	Suite Next reboot
FoundationSuiteK9 securityk9 appxk9	None	Smart License	None
AdvUCSuiteK9 uck9 cme-srst cube	AdvUCSuiteK9	Smart License	AdvUCSuiteK9

#### Technology Package License Information:

Technology	Technology-package Current	Type	Technology-package Next reboot
appxk9	appxk9	Smart License	appxk9
uck9	None	Smart License	None
securityk9	securityk9	Smart License	securityk9
ipbase	ipbasek9	Smart License	ipbasek9

The current throughput level is 50000 kbps

cisco ISR4321/K9 (1RU) processor with 3743802K/3071K bytes of memory.  
 Processor board ID FLM2201V1AX  
 2 Gigabit Ethernet interfaces  
 2 Voice FXS interfaces  
 32768K bytes of non-volatile configuration memory.  
 8388608K bytes of physical memory.  
 7057407K bytes of flash memory at bootflash:.  
 0K bytes of webUI ODM Files at webui:.

Configuration register is 0x2012

COXCME#sh cube status  
 CUBE-Version : 12.7.0  
 SW-Version : 16.12.1a, Platform ISR4321/K9



## Global Cisco Unified CME Settings

In order to enable Cisco Unified CME IP2IP functionality, following command has to be entered:

```
voice service voip
  no ip address trusted authenticate
  mode border-element license capacity 20
  allow-connections sip to sip
  no supplementary-service sip moved-temporarily
  redirect ip2ip
  fax protocol pass-through g711ulaw
  sip
    session refresh
    registrar server expires max 600 min 60
    asserted-id pai
!
```

### Explanation

Command	Description
allow-connections sip to sip	Allow IP2IP connections between two SIP call legs
fax protocol	Specifies the fax protocol
asserted-id	Specifies the privacy header in the outgoing SIP requests and response messages

## Codecs

G711ulaw codec is used for this testing. Codec preferences used to change according to the test plan description.

```
voice class codec 1
  codec preference 1 g711ulaw
  codec preference 2 g711alaw
  codec preference 3 g722-64
!
```



## Dial Peer

### Outbound Dial-peer to Cox:

```
dial-peer voice 100 voip
description Outbound Call to Cox
translation-profile outgoing To-10Dig
destination-pattern .T
session protocol sipv2
session target sip-server
session transport udp
voice-class codec 1
dtmf-relay rtp-nte
no vad
!
```

### Inbound Dial-peer from Cox:

```
dial-peer voice 101 voip
description inbound from Cox
translation-profile incoming to_ext
session protocol sipv2
session transport udp
incoming called-number 402.....
voice-class codec 1
dtmf-relay rtp-nte
no vad
```



## Configuration Example

The following configuration snippet contains a sample configuration of Cisco Unified CME with all parameters mentioned previously.

### Cisco Unified CME:

```
COXCME#sh run
```

```
version 16.12
service timestamps debug datetime msec
service timestamps log datetime msec
service call-home
platform qfp utilization monitor load 80
no platform punt-keepalive disable-kernel-core
!
hostname COXCME
!
boot-start-marker
boot system bootflash:isr4300-universalk9.16.12.01a.SPA.bin
boot-end-marker
!
!
vrf definition Mgmt-intf
!
address-family ipv4
exit-address-family
!
address-family ipv6
exit-address-family
!
enable secret 9
$14$mYRH$jxCCJz1tcdpwoE$Pbxxg7R2/qNJYJCA.1zbFo8gPeDZw93ZgDTB4Rso8yI
!
no aaa new-model
```



call-home

! If contact email address in call-home is configured as sch-smart-licensing@cisco.com

! the email address configured in Cisco Smart License Portal will be used as contact email address to send SCH notifications.

contact-email-addr sch-smart-licensing@cisco.com

profile "CiscoTAC-1"

active

destination transport-method http

no destination transport-method email

!

ip name-server 8.8.8.8

login on-success log

subscriber templating

multilink bundle-name authenticated

!

voice service voip

no ip address trusted authenticate

mode border-element license capacity 20

allow-connections sip to sip

no supplementary-service sip moved-temporarily

redirect ip2ip

fax protocol pass-through g711ulaw

sip

session refresh

registrar server expires max 600 min 60

asserted-id pai

!

voice class codec 1

codec preference 1 g711ulaw

codec preference 2 g711alaw

codec preference 3 g722-64

!

voice class sip-profiles 10



```
request INVITE sip-header Privacy add "Privacy:id"
!
voice class sip-profiles 11
  request INVITE sip-header Diversion add "Diversion:
<sip:6329@10.64.5.142>;privacy=off;reason=unconditional;screen=no"
!
voice register global
  mode cme
  source-address 10.64.5.142 port 2000
  max-dn 20
  max-pool 10
  load 7942 SIP42.9-2-1S
  load 7841 sip78xx.12-0-1-11
  mwi stutter
  tftp-path flash:
  file text
  create profile sync 0013901297582157
  auto-register
!
!
voice register dn 1
  number 6329
  name CoxUser1
  mwi
!
voice register dn 2
  number 6374
  name CoxUser2
!
!
voice register pool 1
  busy-trigger-per-button 1
  id mac 0008.3031.F49B
```



```
type 7975
number 1 dn 1
voice-class codec 1
username 6329 password 1234
call-forward b2bua all 7777
!
voice register pool 2
  busy-trigger-per-button 2
  id mac 2C86.D276.FCC2
  type 7841
  number 1 dn 2
  voice-class codec 1
  username 6374 password 1234
!
!
voice translation-rule 1
  rule 1 /\(^8...\)/ /402932\1/
  rule 2 /\(^6...\)/ /402401\1/
!
voice translation-rule 2
  rule 1 /402932\(...\)/ /\1/
  rule 2 /402401\(...\)/ /\1/
!
!
voice translation-profile To-10Dig
  translate calling 1
!
voice translation-profile to_ext
  translate called 2
!
voice-card 0/1
  no watchdog
!
```



```
voice-card 0/4
  no watchdog
!
no license feature hseck9
license udi pid ISR4321/K9 sn FDO21422BVW
license accept end user agreement
license boot suite AdvUCSuiteK9
license boot level appxk9
license boot level securityk9
memory free low-watermark processor 67123
!
diagnostic bootup level minimal
!
spanning-tree extend system-id

et-analytics
!
redundancy
  mode none
!
interface GigabitEthernet0/0/0
  ip address 10.64.5.142 255.255.0.0
  negotiation auto
!
interface GigabitEthernet0/0/1
  ip address 192.65.x.x 255.255.0.0
  shutdown
  negotiation auto
!
interface Service-Engine0/1/0
  no ip address
!
interface Service-Engine0/4/0
```





```
!  
interface GigabitEthernet0  
  vrf forwarding Mgmt-intf  
  ip address 10.1.1.58 255.255.255.0  
  negotiation auto  
!  
ip forward-protocol nd  
no ip http server  
ip http secure-server  
ip tftp source-interface GigabitEthernet0/0/0  
ip route 0.0.0.0 0.0.0.0 10.64.1.1  
ip route 172.16.0.0 255.255.0.0 10.64.1.1  
ip route 172.17.0.0 255.255.0.0 10.64.1.1  
!  
tftp-server flash:kern2.78xx.12-0-1-11.sbn  
tftp-server flash:kern78xx.12-0-1-11.sbn  
tftp-server flash:rootfs2.78xx.12-0-1-11.sbn  
tftp-server flash:rootfs78xx.12-0-1-11.sbn  
tftp-server flash:sboot2.78xx.12-0-1-11.sbn  
tftp-server flash:sboot78xx.12-0-1-11.sbn  
tftp-server flash:sip78xx.12-0-1-11.loads  
tftp-server flash:apps42.9-2-1TH1-13.sbn  
tftp-server flash:cnu42.9-2-1TH1-13.sbn  
tftp-server flash:cvm42sip.9-2-1TH1-13.sbn  
tftp-server flash:dsp42.9-2-1TH1-13.sbn  
tftp-server flash:jar42sip.9-2-1TH1-13.sbn  
tftp-server flash:SIP42.9-2-1S.loads  
tftp-server flash:term42.default.loads  
tftp-server flash:term62.default.loads  
tftp-server flash:music-on-hold.wav  
!  
control-plane  
!
```



```
voice-port 0/1/0
  cptone IN
  station-id number 4024016374
  caller-id enable
!
voice-port 0/1/1
  cptone IN
  station-id number 4029328367
  caller-id enable
!
mgcp behavior rsip-range tgcp-only
mgcp behavior comedia-role none
mgcp behavior comedia-check-media-src disable
mgcp behavior comedia-sdp-force disable
!
mgcp profile default
!
telephony-service
  no privacy
  conference transfer-pattern
  max-ephones 20
  max-dn 20
  ip source-address 10.64.5.142 port 2000
  time-zone 39
  voicemail 7777
  max-conferences 8 gain -6
  call-forward pattern .T
  moh enable-g711 "flash:music-on-hold.wav"
  transfer-system full-consult
  transfer-pattern .T
  create cnf-files version-stamp 7960 May 19 2020 12:02:00
!
!
```



```
dial-peer voice 100 voip
  description Outbound Call to Cox
  translation-profile outgoing To-10Dig
  destination-pattern .T
  session protocol sipv2
  session target sip-server
  session transport udp
  voice-class codec 1
  dtmf-relay rtp-nte
  no vad
```

!

```
dial-peer voice 101 voip
  description inbound from Cox
  translation-profile incoming to_ext
  session protocol sipv2
  session transport udp
  incoming called-number 402.....
  voice-class codec 1
  dtmf-relay rtp-nte
  no vad
```

!

```
dial-peer voice 102 voip
  description towards CUC
  destination-pattern 7777
  session protocol sipv2
  session target ipv4:10.71.4.15:5060
  session transport udp
  voice-class codec 1
  voice-class sip profiles 11
  dtmf-relay rtp-nte
  no vad
```

!

```
dial-peer voice 103 pots
```



```
preference 2
service session
destination-pattern 8367
no digit-strip
port 0/1/1
!
dial-peer voice 104 pots
preference 2
service session
destination-pattern 6374
no digit-strip
port 0/1/0
!
sip-ua
credentials number 402506**** username 402506**** password 7
022E3D0E093F2D116F443F2F5C realm cox
authentication username 402506**** password 7 022E3D0E093F2D116F443F2F5C
realm cox
mwi-server ipv4:10.71.4.15 expires 3600 port 5060 transport udp unsolicited
registrar ipv4:10.64.4.166 expires 3600
sip-server ipv4:10.64.4.166:5060
!
ephone-dn 1 dual-line
number 6459
label 6459
!
ephone-dn 2 octo-line
number 6476
label 6476
caller-id block
!
ephone 1
device-security-mode none
mac-address CCAF.78D2.3E6D
```



```
username "6459" password 1234
type CIPC
button 1:1
!
ephone 2
privacy on
device-security-mode none
mac-address OCD9.9690.038C
username "6476" password 1234
type 7942
button 1:2
!
line con 0
stopbits 1
line aux 0
stopbits 1
line vty 0 4
exec-timeout 0 0
password *****
login
!
ntp server 10.10.10.5
!
end

COXCME#
```



## Cisco Unity Connection

### Cisco Unity Connection Version



Figure 2: Cisco Unity Connection Version

## Cisco Unity Connection User Configuration

Navigation: Cisco Unity Connection → Users → Users

- Set Alias= **6329** is used for this example
- Set First Name = **cisco** is used to identify this User
- Set Last Name = **cisco** is used for this example
- Set Display Name = **6329** is used in this example
- Set SMTP Address = **6329** is used in this example
- Set Extension = **6329** is used in this example
- Set Phone System = **Cox\_CME** is used in this example



The screenshot shows the Cisco Unity Connection Administration interface. On the left, a navigation tree is expanded to 'Users'. The main content area displays the configuration for a user named 'cisco'. Fields for 'Alias\*', 'First Name', 'Last Name', 'Display Name', and 'SMTP Address' are highlighted with red boxes. The 'SMTP Address' field contains '6329@unity-unity2.lab.tekvizion.com'. Below these fields, the 'LDAP Integration Status' section shows 'Do Not Integrate with LDAP Directory' selected. The 'Phone' section shows 'Extension\*' set to '6329' and 'Phone System' set to 'Cox\_CME'. Other fields like 'Initials', 'Title', 'Employee ID', 'Cross-Server Transfer Extension or URI', 'Outgoing Fax Number', 'Outgoing Fax Server', 'Partition', 'Search Scope', and 'Phone System' are also visible.

Figure 3: Cisco Unity Connection User Configuration

- All other values are set to default

The screenshot shows the continuation of the Cisco Unity Connection Administration interface. The navigation tree on the left is expanded to 'Class of Service'. The main content area displays the configuration for a user named 'cisco'. Fields for 'Class of Service' (set to 'Voice Mail User COS') and 'Active Schedule' (set to 'Weekdays') are highlighted with red boxes. Below these fields, the 'Location' section shows fields for 'Address', 'Building', 'City', 'State', 'Postal Code', and 'Country' (set to 'United States'). The 'Time Zone' is set to '(GMT-06:00) America/Chicago' and the 'Language' is set to 'English(United States)'. Other fields like 'Set for Self-enrollment at Next Sign-In', 'List in Directory', 'Send Non-Delivery Receipts on Failed Message Delivery', 'Skip PIN When Calling From a Known Extension', 'Use Short Calendar Caching Poll Interval', 'Recorded Name', 'Department', and 'Manager' are also visible.

Figure 4: Cisco Unity Connection User Configuration (Cont.)



## Cisco Unity Connection Telephony Integration

Navigation: Telephony Integrations → Phone system

- Set Phone System Name = **Cox\_CME** is used for this example

The screenshot displays the Cisco Unity Connection Administration web interface. The left-hand navigation pane shows the 'Cisco Unity Connection' menu with 'Telephony Integrations' expanded, and 'Phone System' selected. The main content area is titled 'Phone System' and contains several configuration sections:

- Phone System**: The 'Phone System Name\*' field is populated with 'Cox\_CME'. Below it is an unchecked checkbox for 'Default TRAP Phone System'.
- Message Waiting Indicators**: Includes checkboxes for 'Send Message Counts', 'Use Same Port for Enabling and Disabling MWIs', and 'Force All MWIs Off for this Phone System'. A 'Run' button is next to the 'Synchronize All MWIs on This Phone System' option.
- Call Loop Detection by Using DTMF**: Includes an unchecked checkbox for 'Enable for Supervised Transfers' and a checked checkbox for 'Enable for Forwarded Message Notification Calls (by Using DTMF)'. Below these are dropdowns for 'DTMF Tone To Use' (set to 'A') and a 'Guard Time' field (set to '2500' milliseconds).
- Call Loop Detection by Using Extension**: Includes a checked checkbox for 'Enable for Forwarded Message Notification Calls (by Using Extension)'.
- Phone View Settings**: Includes an unchecked checkbox for 'Enable Phone View' and two empty text fields for 'CTI Phone Access Username' and 'CTI Phone Access Password'.
- Outgoing Call Restrictions**: Includes radio buttons for 'Enable outgoing calls' (selected), 'Disable all outgoing calls immediately', and 'Disable all outgoing calls between'. Below are time selection fields for 'Beginning Time' (12:00 AM) and 'Ending Time' (12:00 AM).

At the bottom of the form, there are four buttons: 'Save', 'Delete', 'Previous', and 'Next'. The 'Save' button is highlighted with a red box. A note at the very bottom states: 'Fields marked with an asterisk (\*) are required.'

Figure 5: Cisco Unity Connection Telephony Integration





## Port Group

**Navigation:** Telephony Integration → Port Group

- Set Display Name = **Cox\_CME-1** is used for this example
- Check **Register with SIP server**

**Cisco Unity Connection Administration**  
For Cisco Unified Communications Solutions

Navigation: Cisco Unity Connection Administration | Go  
admin | Search Documentation | About | Sign Out

**Cisco Unity Connection**

- Service Parameters
- Plugins
- Fax Server
- LDAP
  - SAML Single Sign on
  - Authz Servers
  - Cross-Origin Resource Sharing (CC)
- SMTP Configuration
- Advanced
- Telephony Integrations**
  - Phone System
  - Port Group**
  - Port
  - Speech Connect Port
  - Trunk
- Security
- Tools
  - Task Management
  - Bulk Administration Tool
  - Custom Keypad Mapping
  - Migration Utilities
  - Grammar Statistics
  - SMTP Address Search

**Port Group**

Display Name\* Cox\_CME-1

Integration Method SIP

Reset Status Reset Not Required

**Session Initiation Protocol (SIP) Settings**

☒ Register with SIP Server

☐ Authenticate with SIP Server

Authentication Username

Authentication Password

Contact Line Name

SIP Security Profile 5060

SIP Transport Protocol UDP

**Advertised Codec Settings**

Display Name	Packet Size
G.711 mu-law	20
G.729	20

**Message Waiting Indicator Settings**

☒ Enable Message Waiting Indicators

Delay between Requests 0 milliseconds

Maximum Concurrent Requests 0

Retries After Successful Attempt 0

Retry Interval After Successful Attempt 5 milliseconds

Fields marked with an asterisk (\*) are required.

Figure 6: Port Group

**Navigation:** Telephony Integration → Port Group → Edit → Servers

- Set IPv4 Address or Host Name = **10.64.5.142**, Cisco unified CME's IP address is used for this example
- Set Port = **5060**
- Click on **Save**



**Cisco Unity Connection Administration**  
For Cisco Unified Communications Solutions

Navigation: Cisco Unity Connection Administration ▾ Go  
admin | Search Documentation | About | Sign Out

**Cisco Unity Connection**

- Subject Line Formats
- Attachment Descriptions
- Enterprise Parameters
- Service Parameters
- Plugins
- Fax Server
- LDAP
- SAML Single Sign on
- Authz Servers
- Cross-Origin Resource Sharing (CORS)
- SMTP Configuration
- Advanced
- Telephony Integrations
  - Phone System
  - Port Group**
  - Port
  - Speech Connect Port
  - Trunk
- Security
- Tools
  - Task Management
  - Bulk Administration Tool
  - Custom Keypad Mapping
- Migration Utilities
  - Grammar Statistics
  - SMTP Address Search
  - Show Dependencies

**Edit Servers**

Search Port Groups ▸ Port Group Basics (Cox\_CME-1) ▸ Edit Servers

Related Links: Check Telephony Configuration ▾ Go

Port Group Edit Refresh Help

Save

**SIP Servers**

Delete Selected Add

	Order	IPv4 Address or Host Name	IPv6 Address or Host Name	Port	TLS Port
<input type="checkbox"/>	0	10.64.5.142		5060	5061

Delete Selected Add

**TFTP Servers**

Delete Selected Add

**IPv6 Addressing Mode**

Preference for Signaling: IPv4 ▾

Preference for Media: IPv4 ▾

Figure 7: Port Group (Cont.)

## Port

**Navigation:** Telephony Integration → Port

- Set Port Name = **Cox\_CME-1-001** is used for this example
- Set Phone System = **Cox\_CME**
- Set Port Group = **Cox\_CME-1**
- Set Server = **unity-unity2.lab.tekvizion.com** is used for this example



**Cisco Unity Connection Administration**  
For Cisco Unified Communications Solutions

Navigation: Cisco Unity Connection Administration ▼ Go  
admin | Search Documentation | About | Sign Out

**Cisco Unity Connection**

- Subject Line Formats
- Attachment Descriptions
- Enterprise Parameters
- Service Parameters
- Plugins
- Fax Server
- LDAP
- SAML Single Sign on
- Authz Servers
- Cross-Origin Resource Sharing (CORS)
- SMTP Configuration
- Advanced
- Telephony Integrations**
  - Phone System
  - Port Group
  - Port**
  - Speech Connect Port
  - Trunk
  - Security
- Tools
  - Task Management
  - Bulk Administration Tool
  - Custom Keypad Mapping
  - Migration Utilities
  - Grammar Statistics
  - SMTP Address Search
  - Show Dependencies

**Port Basics (Cox\_CME-1-001)**

Search Ports ► Port Basics (Cox\_CME-1-001)  
Related Links: Check Telephony Configuration ▼ Go

Port Refresh Help

Save Delete Previous Next

**Phone System Port**

☒ Enabled

Port Name: Cox\_CME-1-001 Restart

Phone System: Cox\_CME

Port Group: Cox\_CME-1

Server: unity-unity2.lab.tekvizion.com ▼

**Port Behavior**

☒ Answer Calls

☒ Perform Message Notification

☒ Send MWI Requests (may also be disabled by the port group)

☒ Allow TRAP Connections

Save Delete Previous Next

Figure 8: Port

## Acronyms



Acronym	Definitions
CME	Communications Manager Express
CPE	Customer Premise Equipment
CUC	Cisco Unity Connection
DTMF	Dual Tone Multi-Frequency
eSBC	Enterprise Session Border Controller
HA	High Availability
PSTN	Public Switched Telephone Network
SCCP	Skinny Client Control Protocol
SG3	Super Group 3
SIP	Session Initiation Protocol
SP	Service Provider



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