

# What You Make Possible



# Communications Manager for Video Call Control

BRKUCC-2665

# The History



# The History



# Upgrading to TC6

- TC6 has two (2) .cop files for CUCM9.0.
  - Firmware/Configuration
- TC6 keys format are not supported in TC<5.1

# Upgrading to TE6

- If upgrading from TC4.x or lower:
  - Upload sw file
  - Enter keyor
  - Upgrade to TC5
  - Enter key
  - Upgrade to TE6
- Unified .cop file for TE6

# Provisioning

- Leverage TMS and TMS PE to manage release keys and automate software upgrades
- Purge endpoint from TMS post upgrade

TE 4.1

E20

TE6

EX60  
EX90

CUCM only

TC5.x

Split

TC6

Merge

Profile-series  
MX-series  
Quickset  
C-series

EX/MX/C -  
Quickset -  
Profile -Series

# TC/TE software parities and imparities

- Feature parity with TC5.1.4.
- H323 and multiway is disabled
- No feature parity with TC6.
  - CTMS Encrypted calls
  - ISDN Link automatic pairing
  - Web Diagnostics
  - Web GUI and Touch UI differences
  - Diagnostics
  - Multisite downspeeding

# Common Endpoint Capabilities

- SIP URI dialling when registered to CUCM
- Encryption in CUCM Environment
- Ad hoc conferencing in CUCM
- Mediatrace
- CTI/JTAPI Support

# TE6.0 Capabilities

- Voice mail support and message waiting indication.
- CUCM Shared Lines support
- CUCM Call Forward All

# Registering the endpoint

## Device Information

Registration	Registered with Cisco Unified Communications Manager 10.68.37.86
IP Address	<a href="#">10.75.209.39</a>
Active Load ID	TC5.1.1.288225
Download Status	Unknown
<input checked="" type="checkbox"/> Device is Active	
<input checked="" type="checkbox"/> Device is trusted	
MAC Address*	0050600646FB
Description	BJN Medianet EX60
Device Pool*	dp_apcbc-bjn <a href="#">View Details</a>
Common Device Configuration	< None > <a href="#">View Details</a>
Phone Button Template*	Standard Cisco TelePresence EX60

## Protocol Specific Information

BLF Presence Group*	Standard Presence group	<a href="#">Find</a>
MTP Preferred Originating Codec*	711ulaw	
Device Security Profile*	Cisco TelePresence EX60 - Standard SIP Non-Secu	
Rerouting Calling Search Space	css_phone-bjn	
SUBSCRIBE Calling Search Space	css_phone-bjn	
SIP Profile*	Standard SIP Profile	
Digest User	71056001	
<input type="checkbox"/> Media Termination Point Required		
<input type="checkbox"/> Unattended Port		
<input type="checkbox"/> Require DTMF Reception		

**Product Specific Configuration Layout**



Room Name (from Exchange(R))

Web Access\*

SSH Access\*

Default Call Protocol\*

Quality Improvement Server

**Admin username and password**

Admin Username

Admin Password

**Directory Number Information**

Directory Number\*

Route Partition

Description

Alerting Name

ASCII Alerting Name

Allow Control of Device from CTI

Associated Devices

v ^

Dissociate Devices

[Edit Device](#)  
[Edit Line Appearance](#)



# SIP URI dialling on CUCM

- EX90 and EX60 Can now register natively to CUCM with a URI such as 'example@cisco.com'
- Requires CUCM 9.0 or later
- Earlier CUCM versions requires numbers only (E164).
- Provisioning of alphanumeric URI is not supported in CUCM 9.0
  - DN Provisioning only via CUCM
  - Alphanumeric manually configured on endpoint

# SIP URI Advantages

- VCS integration
- 3rd party integration
- AD integration

# SIP URI Dialling

Case sensitive

myFineUri@cisco.com

- In CUCM 9.0 URIs are aliases for directory numbers (DN)
- DN registration is the main endpoint URI
  - Cisco Unified CM Administration->Device->Phone-><DEVICE>->Line [x]

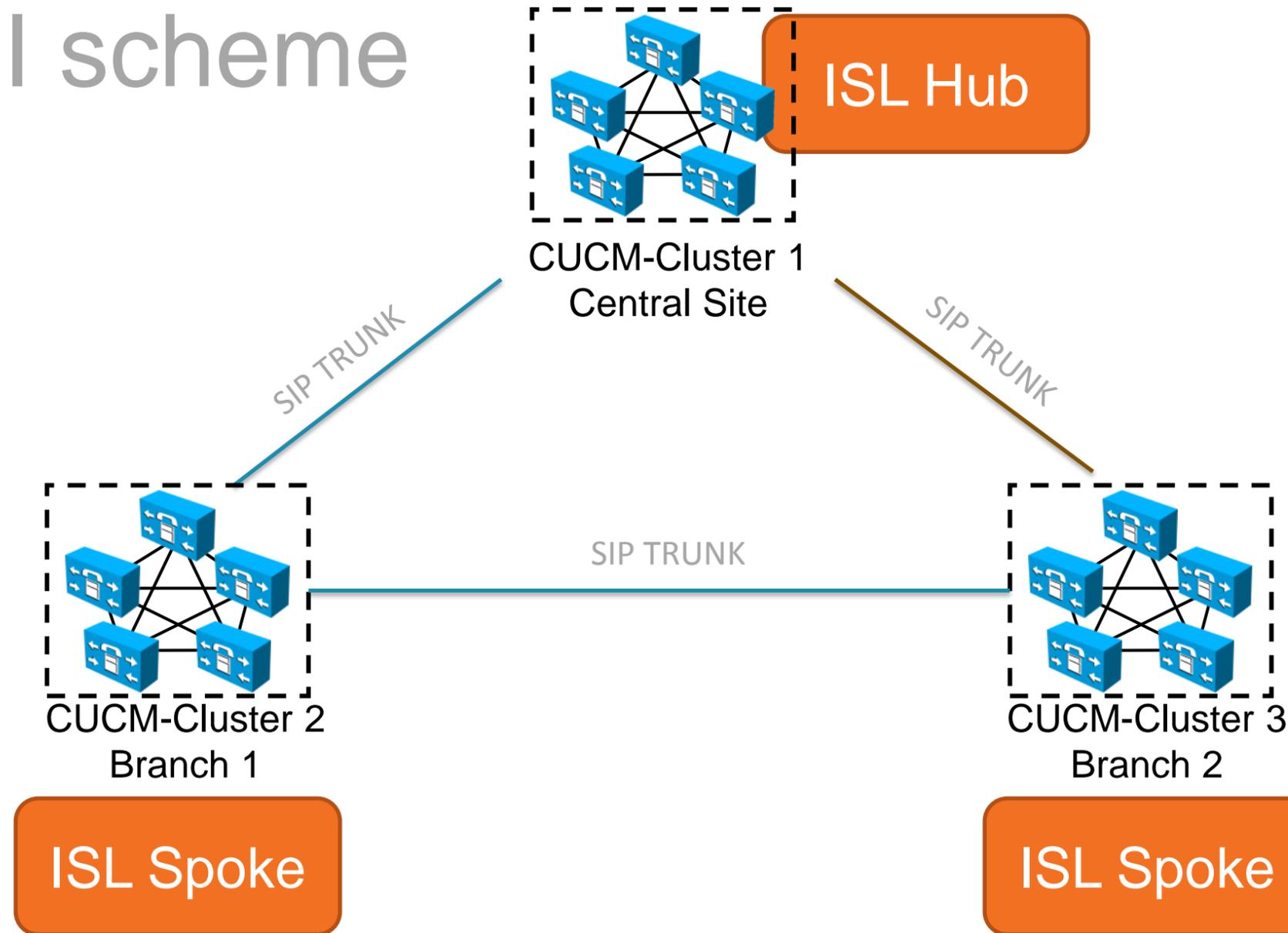
Primary	URI	Partition	Edit/Remove
<input checked="" type="checkbox"/>	malin@tipbu.com <input type="text" value="malin@tipbu.com"/>	Directory URI <input type="text" value=" &lt; None &gt;"/>	<a href="#">Edit End User</a> <input type="button" value=""/>

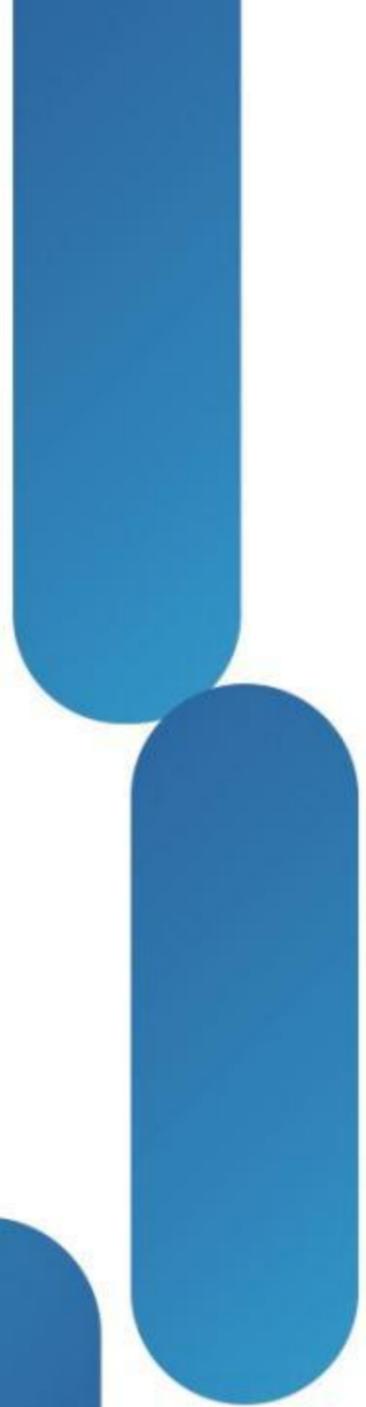
# Support for URI Dialling

- URI calls between clusters can't be routed using a prefix
- Multiple clusters can have the same domain suffix
  - No suffix based routing decisions
- Inter-Cluster Lookup Services (ILS) helps replicate respective catalog of registered URI
- ILS maps a URI to a cluster URI Route String.
  - Route the call to a SIP trunk using a SIP Route Pattern that matches the Cluster URI Route String.

# Support for URI Dialling

Flat URI scheme case





# CUCM Encryption Refresher

- **CAPF (Certificate Authority Proxy Function):**

- Process by which supported devices can request locally significant certificates by using Cisco Unified Communications Manager Administration. This service runs normally in the publisher. It can also work as proxy to import certificates created by a different Certificate Authority (CA)

- **LSC (Locally Significant Certificates):**

- A digital X.509v3 certificate that CAPF issues; installed on the end-point or JTAPI/TAPI/CTI application

## ■ **CTL (Certificate Trust List):**

- A file, which is created with the CTL Client (software installed on windows machine) and signed by the Cisco Site Administrator Security Token (USB token, order separately), that contains a list of certificates for servers that the end-point is to trust.

## ■ **sRTP (Secure Real-Time Transport Protocol):**

- Secure Real-Time Transport Protocol that secures voice conversation in the network and provides protection against replay attacks.

## ■ **TLS (Transport Layer Security):**

- Cryptographic protocol that provides secure and reliable data transfer between two systems or devices, by using secure ports and certificate exchange. TLS secures and controls connections among Cisco Unified Communications Manager-controlled systems, and devices. Functionally equivalent to SSL.

## ■ **ITL (Identity Trust List):**

- Introduced in CUCM8.x onwards, contains the minimum list of certificates that are required by the end-point for authentication, decryption of Phone Configuration file and contacting the TVS service

# CUCM Encryption

- The EX90 and EX60 now support encryption when registered to CUCM.
- Requires that CUCM security mode is installed and configured

## Any version

VCS



sRTP

## Up to TE/TC 5.1

CUCM



No encryption available

## From TE 6.0

CUCM



CAPF/CTL, TLS/sRTP

# Encryption in CUCM Environment

- CAPF using CTL
- Secure Real Time Protocol
- End to end security and trust
- No Configuration encryption

## Room TelePresence

Encryption is showed on the screen (OSD)



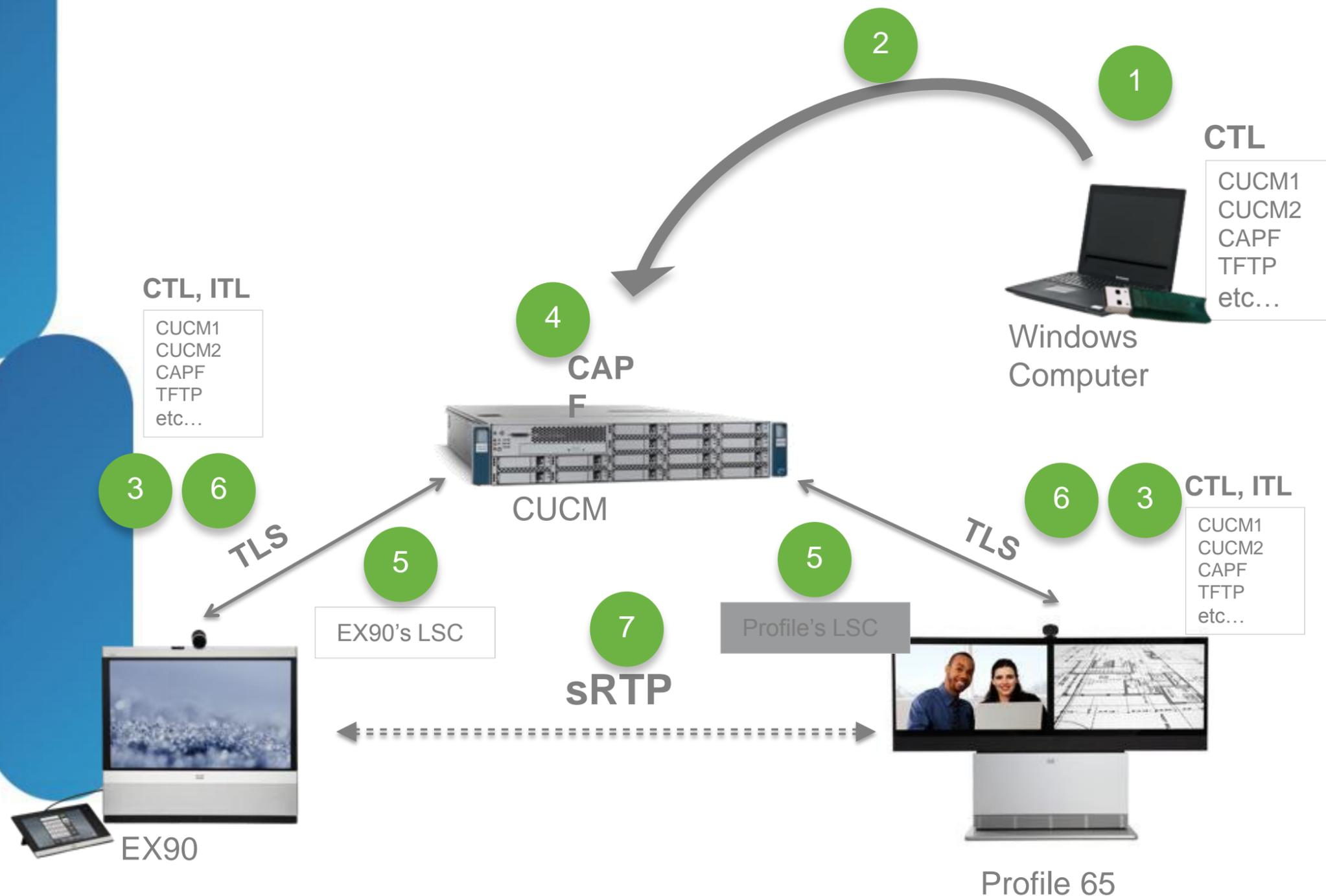
## Personal TelePresence

Encryption is shown on the Touch UI



## Main Benefit

A consistent encryption behaviour on all CUCM Registered Endpoints



1. Admin enables CAPF, and CTL provider on CUCM Cluster
2. Admin runs CTL app. CTL is created in CUCM and security mode set to mixed for the cluster
3. CTL, ITL and configuration are downloaded to end-point from TFTP while booting-up. Communication initiated by end-point
4. Each end-point uses CAPF Client to request a LSC from CUCM CAPF – end-point uses self signed certs – \*
5. LSC downloaded to each end-point\*
6. Secured communication is established between end-point and CUCM over TLS now using the LSC
7. sRTP between end-points is now possible

**Note:** Configuration encryption and MIC are not available in TC/TE 6

- When to delete the CTL/ITL?
  - Changing the CUCM IP address, or host name
  - Moving the end-point between CUCM clusters
  - Others (basically, anything that means re-generating/changing the CUCM certificate)
  - Security By Default (SBD CUCM >8.x) may cause troubles when getting the devices into secure mode the first time. In that case, delete the CTL from the end-point and try again
  
- You can always delete the CTL/ITL by going into the end-point configuration in the security section or by re-running the provisioning wizard\*

# Configuring Security CUCM 9/TC6,TE6

- The CTL must be created. For this use the CTL plug-in in CM Administration->Application->Plugins

The screenshot shows the Cisco Unified CM Administration web interface. The navigation menu is open to 'Application', and the 'Plugins' sub-menu is selected. Below the menu, a status bar indicates '13 records found'. A search filter is applied: 'Name begins with' and 'Plugin Type equals Installation'. The table below lists the plugins, with the 'Cisco CTL Client' entry highlighted by a red box.

	Plugin Name	Description
<a href="#">Download</a>	<a href="#">Cisco AXL Toolkit</a>	Cisco Administrative XML (AXL) Toolkit enables Developers to create applications that create, read, update and delete provisioning objects on the Cisco Unified Communications Manager Publisher. The zip file contains Java-based libraries that use SOAP over HTTP/HTTPS to send and receive AXL requests and responses. Install this toolkit on Developer workstations where AXL applications will be developed. SHA1(/usr/local/thirdparty/jakarta-tomcat/webapps/plugins/axlsqltoolkit.zip)= 26:e6:6b:61:2e:05:88:9b:d2:57:e4:f8:88:27:4f:4d:aa:ac:7b:1e
<a href="#">Download</a>	<a href="#">Cisco CTL Client</a>	Install the Cisco Certificate Trust List (CTL) client to digitally sign certificates stored on the TFTP server. The client retrieves the CTL file from the Cisco TFTP server, digitally signs the CTL file using a security token and then updates the file on the Cisco TFTP server. Install this plug-in on Windows 32-bit or Windows 64-bit operating system computers. SHA1(/usr/local/thirdparty/jakarta-tomcat/webapps/plugins/CiscoCTLClient.exe)= cd:9f:82:46:b9:f1:da:35:5d:58:41:bb:61:78:23:38:cd:ae:1b:15

- Check mixed mode

- CM Admin → System → Enterprise Parameters → Security Parameters.

The screenshot shows the Cisco Unified CM Administration web interface. The navigation menu at the top includes 'System', 'Call Routing', 'Media Resources', 'Advanced Features', 'Device', 'Application', 'User Management', 'Bulk Administration', and 'Help'. The 'User Management' menu is highlighted with a red box. In the left sidebar, 'Enterprise Parameters' is selected and highlighted with a red box. The 'Security Parameters' section is expanded, showing a table of configuration items:

Security Parameters	
Cluster Security Mode *	1
CAPF Phone Port *	3804
CAPF Operation Expires in (days) *	10
Enable Caching *	False

- At this point any end-point connecting to the CUCM should be able to download the CTL/ITL and it's configuration.

The best way to check if this is happening , is by looking at the logs in the end-point. The log file to look into is the application.log

The screenshot shows the Cisco TelePresence MX200 web interface. The 'Diagnostics' menu is selected, and 'Log Files' is highlighted. The log output shows several entries related to downloading trustlists and provisioning. Red boxes and arrows highlight specific log lines:

- CTL**: Points to the log line: `TRUSTLIST I: Download of trustlist 'http://10.1.100.12:6970/CTLSEP005060069bcf.tlv' complete`
- ITL**: Points to the log line: `TRUSTLIST I: Download of trustlist 'http://10.1.100.12:6970/ITLSEP005060069bcf.tlv' complete`
- Configuration**: Points to the log line: `PROV I: void CUCMProvision::CUCMProvisionUser::configItem() 'http://10.1.100.12:6970/SEP005060069bcf.cnf.xml.sgn'`
- Another red box highlights the final log line: `PROV I: void CUCMProvision::CUCMProvisionUser::report(bool) Provisioned Successfully`

- The LSC must be created by CAPF and sent to each end-point.
  - CM Administration → Device → SELECTED-ENDPOINT → CAPF Information

The screenshot shows the Cisco Unified CM Administration web interface. The navigation path is: System > Call Routing > Media Resources > Advanced Features > Device > Application > User Management > Bulk Administration > Help. The 'Device' menu is open, and 'Phone' is selected. The main content area shows the configuration for a Cisco TelePresence MX200 device. The 'Certification Authority Proxy Function (CAPF) Information' section is highlighted, showing the following settings:

- Certificate Operation\*: Install/Upgrade
- Authentication Mode\*: By Null String
- Authentication String: 12345
- Key Size (Bits)\*: 1024
- Operation Completes By: 2012 7 28 12 (YYYY:MM:DD:HH)

Red arrows point to the 'Install/Upgrade' dropdown, the 'By Null String' dropdown, and the 'Authentication String' input field.

Select Install/Upgrade

At the time of launching TE/TC 6, there is no MIC on these devices, therefore you can select only between *Null String*, or *Authentication String*

If *By Authentication String* has been chosen, you have to input a string here, and it should match the one to input in the end-point\*

# Troubleshooting CUCM Encryption

- Successful verification via web UI, Touch device or CLI

## Cisco Call Manager (CUCM) - Certification Authority Proxy Function (CAPF) Information

CUCM status	CUCM is enabled.
CTL status	CTL is installed.
ITL status	ITL is installed.
LSC status	Certificates are installed.
Operation status	No pending operations...

Delete CTL/ITL

- CUCM encrypted status can be checked using the 'xstatus Provisioning CUCM xAPI':
  - \*s Provisioning CUCM CAPF Mode: IgnoreAuth
  - \*s Provisioning CUCM CAPF ServerName: "assip-cucm-3.rd.tandberg.com"
  - \*s Provisioning CUCM CAPF ServerPort: 3804
  - \*s Provisioning CUCM CAPF LSC: Installed
  - \*s Provisioning CUCM CAPF OperationState: NonPending
  - \*s Provisioning CUCM CAPF OperationResult: CAPFLSCUpdated
  - \*s Provisioning CUCM ProvisionSecurity: Signed
  - \*s Provisioning CUCM CTL State: Installed
- If LSC / CTL state is NOT 'Installed' the system is not able to connect securely to the CUCM

- Create the secured profiles

The screenshot displays the Cisco Unified CM Administration interface. The 'System' menu is expanded, and the path 'Security > Phone Security Profile' is highlighted. The configuration form for a 'Phone Security Profile' is shown with the following details:

- Phone Security Profile Information:**
  - Product Type: Cisco TelePresence MX200
  - Device Protocol: SIP
  - Name\*: Cisco TelePresence MX200 - Standard SIP Secure Prof
  - Description: Cisco TelePresence MX200 - Standard SIP Secure Prof
  - Nonce Validity Time\*: 600
  - Device Security Mode: Encrypted
  - Transport Type\*: TLS
  - Enable Digest Authentication
  - Exclude Digest Credentials in Configuration File
- Phone Security Profile CAPF Information:**
  - Authentication Mode\*: By Null String
  - Key Size (Bits)\*: 1024
  - Note: These fields are related to the CAPF Information settings on the Phone Configuration page.
- Parameters used in Phone:**
  - SIP Phone Port\*: 5060

At the bottom of the form, there are buttons for 'Save', 'Delete', 'Copy', 'Reset', 'Apply Config', and 'Add New'.

Set to Encrypted

Set to TLS

Set to what was selected on the CAP Communication (Slide 13)

- Apply the secured profile to each device

The screenshot displays the Cisco Unified CM Administration web interface. At the top, the navigation bar includes the Cisco logo, the title "Cisco Unified CM Administration For Cisco Unified Communications Solutions", and a navigation dropdown menu set to "Cisco Unified CM Administration" with a "Go" button. Below this, a secondary navigation bar shows the user role as "administrator" and provides links for "Search Documentation", "About", and "Logout".

The main content area is titled "Phone Configuration" and features a toolbar with icons for "Save", "Delete", "Copy", "Reset", and "Apply Config". A dropdown menu is open, showing options: "CTI Route Point", "Gatekeeper", "Gateway", "Phone" (highlighted in red), "Trunk", "Remote Destination", and "Device Settings".

On the left side, the "Status" section shows "Status: Ready". Below it, the "Association Information" section contains a table with two entries:

Line	Description
1	Line [1] - 7710 (no partition)
2	Line [2] - Add a new DN

The "Device Information" section on the left lists fields for "Registration", "IP Address", "Active Load ID", and "Download Status", with checkboxes for "Device is Active" and "Device is trusted", both of which are checked.

The "Protocol Specific Information" section on the right contains several configuration fields:

- Packet Capture Mode\*: None
- Packet Capture Duration: 0
- BLF Presence Group\*: Standard Presence group
- MTP Preferred Originating Codec\*: 711ulaw
- Device Security Profile\*: Cisco TelePresence MX200 - Standard SIP Secure Prc (highlighted in red)
- Rerouting Calling Search Space: < None >
- SUBSCRIBE Calling Search Space: < None >
- SIP Profile\*: Standard SIP Profile
- Digest User: < None >

At the bottom of the "Protocol Specific Information" section, there are three unchecked checkboxes: "Media Termination Point Required", "Unattended Port", and "Require DTMF Reception".

# Register the endpoint

- If endpoint is registered to CUCM in non-secure mode
  - *xConfiguration Provisioning Mode:off*
  - *xCommand Provisioning CUCM CTL Delete*
  - *xConfiguration Provisioning ExternalManager Address: <dns/ip-address to the CUCM>*
  - *xConfiguration Provisioning Mode:CUCM*

- CUCM and end-point communication Traces before Encryption

Source	Destination	Protocol	Info
10.1.100.53	cucm.tipbu.com	SIP	Request: REGISTER sip:cucm.tipbu.com (application/x-cisco-remotecc-request+xml)
cucm.tipbu.com	10.1.100.53	SIP	Status: 100 Trying (0 bindings)
cucm.tipbu.com	10.1.100.53	SIP	Status: 200 OK (1 bindings)
cucm.tipbu.com	10.1.100.53	SIP	Request: REFER sip:7702@10.1.100.53:59080
cucm.tipbu.com	10.1.100.53	SIP	Request: NOTIFY sip:7702@10.1.100.53:59080 (text/plain)
cucm.tipbu.com	10.1.100.53	SIP	Request: NOTIFY sip:36e6ff87-81d2-c05d-3e98-8496f30ca54d@10.1.100.53:59080;transport=tcp
cucm.tipbu.com	10.1.100.53	SIP	Request: SUBSCRIBE sip:36e6ff87-81d2-c05d-3e98-8496f30ca54d@10.1.100.53:59080
10.1.100.53	cucm.tipbu.com	SIP	Status: 200 OK
10.1.100.53	cucm.tipbu.com	SIP	Status: 481 Subscription does not exist
10.1.100.53	cucm.tipbu.com	SIP	Status: 200 OK
10.1.100.53	cucm.tipbu.com	SIP	Status: 200 OK
10.1.100.53	cucm.tipbu.com	SIP	Request: NOTIFY sip:7702@10.1.100.12:5060;transport=tcp
cucm.tipbu.com	10.1.100.53	SIP	Status: 200 OK
10.1.100.53	cucm.tipbu.com	SIP	Request: SUBSCRIBE sip:7702@cucm.tipbu.com
cucm.tipbu.com	10.1.100.53	SIP	Status: 200 OK
cucm.tipbu.com	10.1.100.53	SIP	Request: NOTIFY sip:36e6ff87-81d2-c05d-3e98-8496f30ca54d@10.1.100.53:59080;transport=tcp
10.1.100.53	cucm.tipbu.com	SIP	Status: 200 OK
10.1.100.53	cucm.tipbu.com	SIP	Request: NOTIFY sip:7702@10.1.100.12:5060;transport=tcp
cucm.tipbu.com	10.1.100.53	SIP	Status: 200 OK

cucm.tipbu.com = 10.1.100.12  
MX200 = 10.1.100.53



- Call between two end-points traces before encryption

10.1.100.51	10.1.100.53	RTP	PT=MP4A-LATM, SSRC=0xAD49C35F, Seq=2341, Time=1713868285
10.1.100.51	10.1.100.53	RTP	PT=MP4A-LATM, SSRC=0xAD49C35F, Seq=2342, Time=1713870085
10.1.100.51	10.1.100.53	RTP	PT=MP4A-LATM, SSRC=0xAD49C35F, Seq=2343, Time=1713871885
10.1.100.51	10.1.100.53	RTP	PT=MP4A-LATM, SSRC=0xAD49C35F, Seq=2344, Time=1713873685
10.1.100.51	10.1.100.53	RTP	PT=MP4A-LATM, SSRC=0xAD49C35F, Seq=2345, Time=1713875485
10.1.100.51	10.1.100.53	RTP	PT=MP4A-LATM, SSRC=0xAD49C35F, Seq=2346, Time=1713877285
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39167, Time=3392998479 NAL unit - Sequence parameter set
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39168, Time=3392998479 NAL unit - Picture parameter set
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39169, Time=3392998479 FU-A
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39170, Time=3392998479 FU-A
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39171, Time=3392998479 FU-A
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39172, Time=3392998479 FU-A
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39173, Time=3392998479 FU-A
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39174, Time=3392998479 FU-A
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39175, Time=3392998479 FU-A
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39176, Time=3392998479 FU-A
10.1.100.51	10.1.100.53	H264	PT=H264, SSRC=0xE363E6C9, Seq=39177, Time=3392998479 FU-A
10.1.100.51	10.1.100.53	RTP	PT=MP4A-LATM, SSRC=0xAD49C35F, Seq=2347, Time=1713879085

EX60 = 10.1.100.51  
MX200 = 10.1.100.53

## ■ CUCM and end-point communication Traces after Encryption

10.1.100.53	cucm.tipbu.com	TCP	39322 > sip-tls [ACK] Seq=1 Ack=1 Win=14624 Len=0 TSval=4294764964 TSecr=38403276
10.1.100.53	cucm.tipbu.com	TLSv1	Client Hello
cucm.tipbu.com	10.1.100.53	TCP	sip-tls > 39322 [ACK] Seq=1 Ack=262 Win=6912 Len=0 TSval=38404088 TSecr=4294765773
cucm.tipbu.com	10.1.100.53	TLSv1	Server Hello, Certificate, Certificate Request, Server Hello Done
10.1.100.53	cucm.tipbu.com	TCP	39322 > sip-tls [ACK] Seq=262 Ack=793 Win=16192 Len=0 TSval=4294765822 TSecr=38404088
10.1.100.53	cucm.tipbu.com	TLSv1	Certificate, Client Key Exchange, Certificate Verify, Change Cipher Spec, Encrypted Handshake Message
cucm.tipbu.com	10.1.100.53	TCP	sip-tls > 39322 [ACK] Seq=793 Ack=1922 Win=12672 Len=0 TSval=38404408 TSecr=4294766093
cucm.tipbu.com	10.1.100.53	TLSv1	New Session Ticket, Change Cipher Spec, Encrypted Handshake Message
10.1.100.53	cucm.tipbu.com	TCP	39322 > sip-tls [ACK] Seq=1922 Ack=1651 Win=17920 Len=0 TSval=4294766101 TSecr=38404412
10.1.100.53	cucm.tipbu.com	TLSv1	Application Data, Application Data
cucm.tipbu.com	10.1.100.53	TCP	sip-tls > 39322 [ACK] Seq=1651 Ack=3468 Win=15616 Len=0 TSval=38404576 TSecr=4294766263
cucm.tipbu.com	10.1.100.53	TLSv1	Application Data, Application Data
10.1.100.53	cucm.tipbu.com	TCP	39322 > sip-tls [ACK] Seq=3468 Ack=2045 Win=19616 Len=0 TSval=4294766265 TSecr=38404576
cucm.tipbu.com	10.1.100.53	TLSv1	Application Data, Application Data
cucm.tipbu.com	10.1.100.53	TLSv1	Application Data, Application Data
cucm.tipbu.com	10.1.100.53	TLSv1	Application Data, Application Data
10.1.100.53	cucm.tipbu.com	TCP	39322 > sip-tls [ACK] Seq=3468 Ack=3127 Win=21792 Len=0 TSval=4294766272 TSecr=38404584
10.1.100.53	cucm.tipbu.com	TCP	39322 > sip-tls [ACK] Seq=3468 Ack=3953 Win=23968 Len=0 TSval=4294766273 TSecr=38404584
10.1.100.53	cucm.tipbu.com	TCP	39322 > sip-tls [ACK] Seq=3468 Ack=4635 Win=26112 Len=0 TSval=4294766273 TSecr=38404584
cucm.tipbu.com	10.1.100.53	TLSv1	Application Data, Application Data
cucm.tipbu.com	10.1.100.53	TLSv1	Application Data, Application Data

cucm.tipbu.com = 10.1.100.12  
MX200 = 10.1.100.53



# Scalable Ad-hoc Conferencing

Seamless escalation  
to multi-party conferencing

Scalable and effective use of  
conferencing resources

Automatic de-escalation from  
multipoint to point-to-point



# Ad-hoc Conferencing

## Endpoint Multisite

- Extra Audio Call
- Embedded Multisite (licence)

## CUCM Media Resource

- Escalate to CUCM MCU
- Dynamic deescalation
- Future platform

## VCS Multiway

- Escalate to VCS MCU
- No deescalation
- Legacy

# UC Manager Ad Hoc Conferencing

- When configured, UCM MCU Ad Hoc Conferencing will be dominant over embedded Multisite
- “As Needed” dynamic escalation TO an MCU or de-escalation FROM an MCU (better port management)

# UCM remote-cc

- Only supported by Cisco SCCP and SIP endpoints using Cisco-proprietary SIP remote-cc extensions
- Allows chains to occur in many circumstances
- Provides telephony-like UE including early-attended, and CTI/3PCC support
- Proper recovery handling and de-escalating back to pt-to-pt
- Centralised (per UCM cluster) policies

# Ad Hoc Conferencing Configuration

- On the MCU, configuring settings

1. Go to **Settings > Conferences**.

2. Configure the fields on the MCU as follows:

MCU Setting	Value	Comment
Media port reservation	Enabled	
Incoming calls to unknown conferences or auto attendants	Disconnect caller	

3. Click **Apply changes**.

## ■ On Unified CM, configuring conference features

To configure conference features such as the maximum number of participants:

1. Go to **System > Service Parameters**.
2. Select the relevant Unified CM Server.
3. Select the **Cisco CallManager (Active)** as the service.
4. Select **Advanced** to show advanced options.
5. Configure the **Clusterwide Parameters (Feature - Conference)** section as required.

## ■ On Unified CM, adding the MCU

Add the MCU to Unified CM as a manageable device as follows:

1. Go to **Media Resources > Conference Bridge**.
2. Click **Add New**.
3. Select **Conference Bridge Type** as **Cisco TelePresence MCU**.
4. Enter relevant fields and click **Save**.

# TelePresence MCU with Unified CM

## Conference Bridge Information

Conference Bridge : mcu0aTTGTME (mcu0aTTGTME)  
Registration Registered with Cisco Unified Communications Manager ttgtmecucm1  
IP Address 172.19.236.207

## MCU Conference Bridge Info

Conference Bridge Type \* Cisco TelePresence MCU  
 Device is trusted  
Conference Bridge Name \* mcu0aTTGTME  
Destination Address \* 172.19.236.207  
Description mcu0aTTGTME  
Device Pool \* Default  
Common Device Configuration < None >  
Location \* Hub\_None  
Use Trusted Relay Point \* Default

## SIP Interface Info

Unified CM SIP Port \* 5060  
MCU Conference Bridge SIP Port \* 5060

## HTTP Interface Info

Username \* admin  
Password \* ●●●●●●●●  
Confirm Password \* ●●●●●●●●  
HTTP Port \* 80

## ■ On Unified CM, configuring a media resource group list

1. Go **Media Resources > Media Resource Group**.
2. Click **Add New**.
3. Choose a name and move the MCU(s) into the **Selected Media Resources** area.
4. Click **Save**.
5. Go to **Media Resources > Media Resource Group List**.
6. Click **Add New**.
7. Choose a name and move the created Media Resource Groups into the **Selected Media Resource Groups** area.
8. Click **Save**.

## ■ On Unified CM, assign a MRGL to a device

1. Go to **Device > Phone**.
2. Select a device.
3. Choose the **Media Resource Group List** that you created earlier.
4. Click **Save**.

- All centralised configuration
- Added as a TelePresence MCU in CUCM
- Set Multipoint settings to <<Use MRGL>>
  - Default for EX90 is «Use Endpoint»
  - Default for EX60 is «Use Media Resource Group List»
- The setup can be verified by running *xstatus Conference UseBuiltInBridge*

# What's Cisco MediaNet – Mediatraces

## Medianet is

An Architecture/Blueprint for successful deployment of multiple media and business critical applications

## Medianet is **not**

A product, SKU, or a marketecture to describe just QoS/CAC

## Mediatraces is

A Cisco IOS feature that discovers the routers and switches along the path of an IP flow, and collects critical information hop by hop on specific media streams as they traverse the network. It's a diagnostic tool similar to traceroute, that should be enabled on each of the network nodes you want to collect information from.

# MediaNet – Mediatraces



1. We're having bad image, what do we do?

TP Room 1  
Room with poor video quality



2. Prime Collaboration Manager is pro-actively informing that TP room 1 is suffering congestion

Administrator in front of PC detecting that there is a problem



Administrator fixing the problem

3. With few clicks administrators can fix the routers issues remotely



4. Oh!, Now it's fixed, and we didn't do anything

Great quality re-established

# MediaNet – Mediatraces



TelePresence

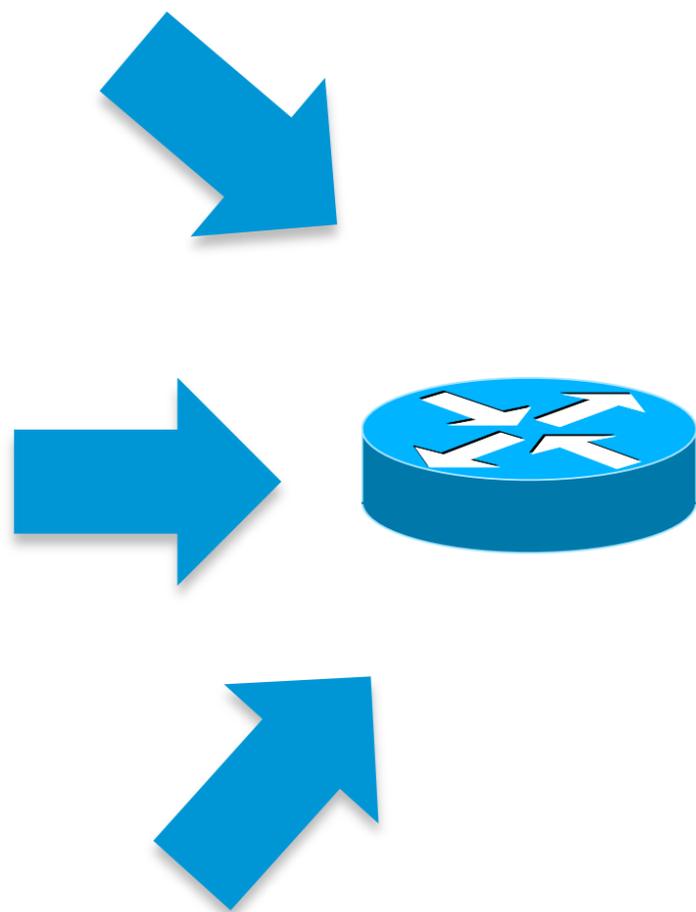


Physical Security



YouTube

HTTP Streams



- Video marked as “TelePresence” at origin
- Monitor only the flows that matters
- Sift through mountains of data
- No deep packet inspection (DPI)
- Works for encrypted calls

# MediaNet – Mediatraces

- Mediatraces information collection methods
  - Exec command to perform on-demand collection of statistics (reactive method)
  - By configuring Cisco Mediatrace to start a recurring monitoring session at a specific time and on specific days (pro-active method)
- Cisco Prime Collaboration is a proactive tool

# Medianet –Metadata

- The Medianet metadata = Application information
- OOB = Encryption friendly
- Leverages RSVP as a transport with a second 5-tuple
  - Follows same path as media

# Medianet – Mediatriace



## **New in TE/TC/TX 6.0:**

TelePresence end-points will include a MediaNet MSI that sends application ID/Globally Unique Session ID as part of the traffic stream and session set up

Media Services Interface (MSI): is a software component that enables end-points to consistently take advantage of intelligent network services to improve the quality of experience and reduce the cost of deployment and operations.

# Display CDP information

## ■ xstat network 1 CDP

- \*s Network 1 CDP Platform: "cisco WS-C3750X-48P"
- \*s Network 1 CDP Version: "Cisco IOS Software, C3750E Software (C3750E-UNIVERSALK9-M), Version 15.0(1)SE2, RELEASE SOFTWARE (fc3)\*Technical Support: <http://www.cisco.com/techsupport>\*Copyright (c) 1986-2011 by Cisco Systems, Inc.\*Compiled Thu 22-Dec-11 00:05 by prod\_rel\_team"
- \*s Network 1 CDP Capabilities: "0x0029"
- \*s Network 1 CDP DeviceId: "XXXX.cisco.com"
- \*s Network 1 CDP PortID: "GigabitEthernet2/0/3"
- \*s Network 1 CDP Duplex: "Full"
- \*s Network 1 CDP VTPMgmtDomain: ""
- \*s Network 1 CDP Address: "X.X.X.X"
- \*s Network 1 CDP PrimaryMgmtAddress: "X.X.X.X"
- \*s Network 1 CDP SysName: ""
- \*s Network 1 CDP SysObjectID: ""
- \*s Network 1 CDP VoIPApplianceVlanID: "300"
- \*\* end

# CTI/JTAPI Support

- EX90 and EX60 support JTAPI
- Cisco Remote expert 1.8 is now supported.

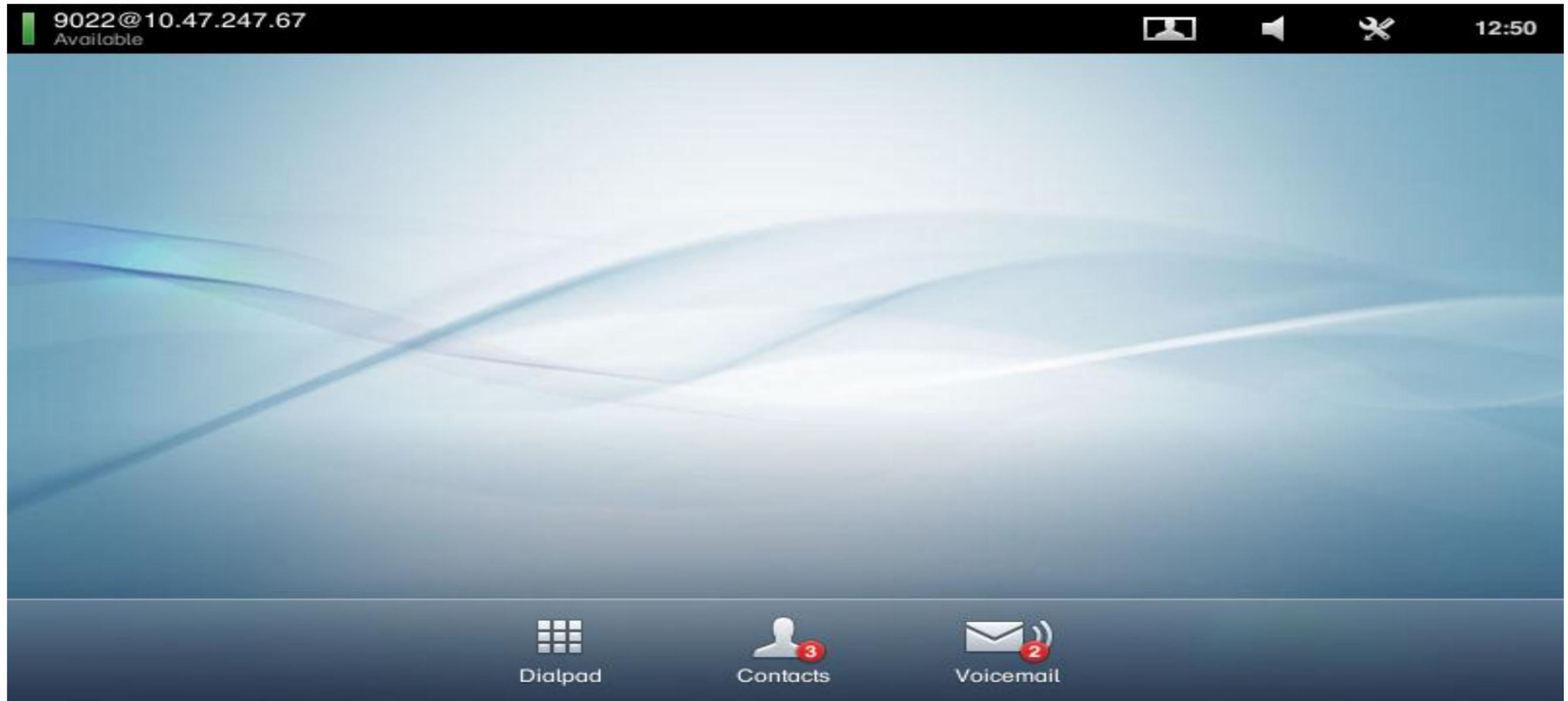
# Remote Expert Support

- CTI in TE6/CUCM 9.0 is limited to the Remote Expert solution
  - CTI Monitoring of device availability
  - CTI Remote Call Control (remote-cc)
- Jabber 9 can control EX Series, however not officially supported
  - CTI basic call features supported in Experimental Mode:
  - Call, Accept, Transfer, Hold, Resume, Click to Call (websites)

# Voicemail

- The EX90 and EX60 support Voicemail
  - *xconfiguration SIP Profile 1 Mailbox: "voicemail@cisco.com"*
- When a mailbox is configured, a Voicemail icon will be displayed on the bottom right side of the main menu.
- Message waiting indicator badge showing number of new messages.
- Only CUCM is supported for message waiting indicator badge

# Voicemail with MWI



# Voicemail Integration Guiding Principles

- Unity can be integrated with UCM, or with VCS, or both (dual integration)
  - *VCS X6.1 supports direct SIP integration with Unity*
  - *VCS X7.0 adds support for receiving and responding to the diversion header sent by UCM when Unity integration is via UCM*
- Unity Voicemail does not support alpha-numeric URIs
  - *Unity voicemail pilot number must be a numeric address*
  - *Subscribers must have numeric addresses and their numeric addresses must be used in the diversion headers sent to Unity*
  - *Callers leaving a message for a subscriber must have a numeric address or else the subscriber will not be able to reply to message or call back the sender*
- Unity supports KPML and RFC 2833 DTMF
  - *VC endpoints and H.320 gateways must support either KPML or RFC 2833*
- Message Waiting Indication (MWI) is supported with TE6

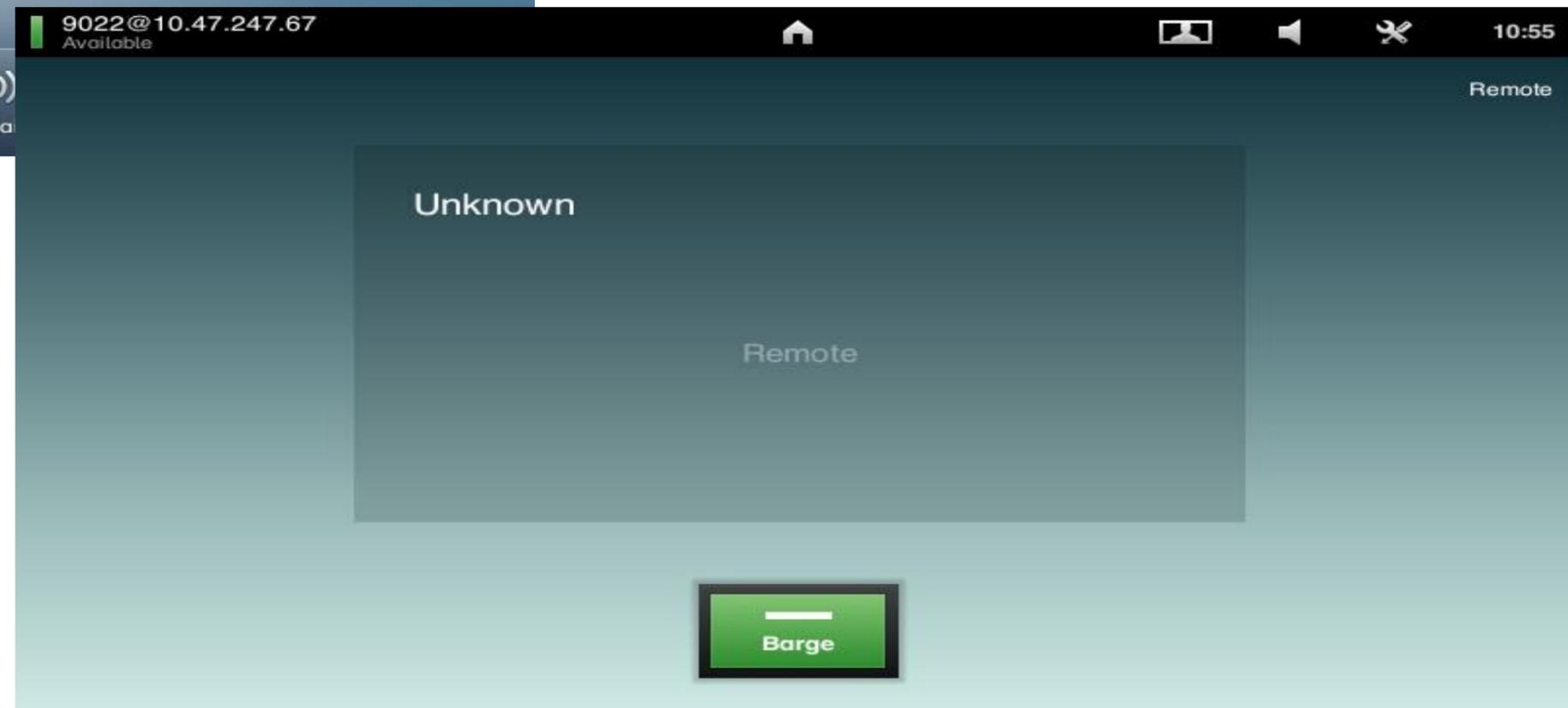
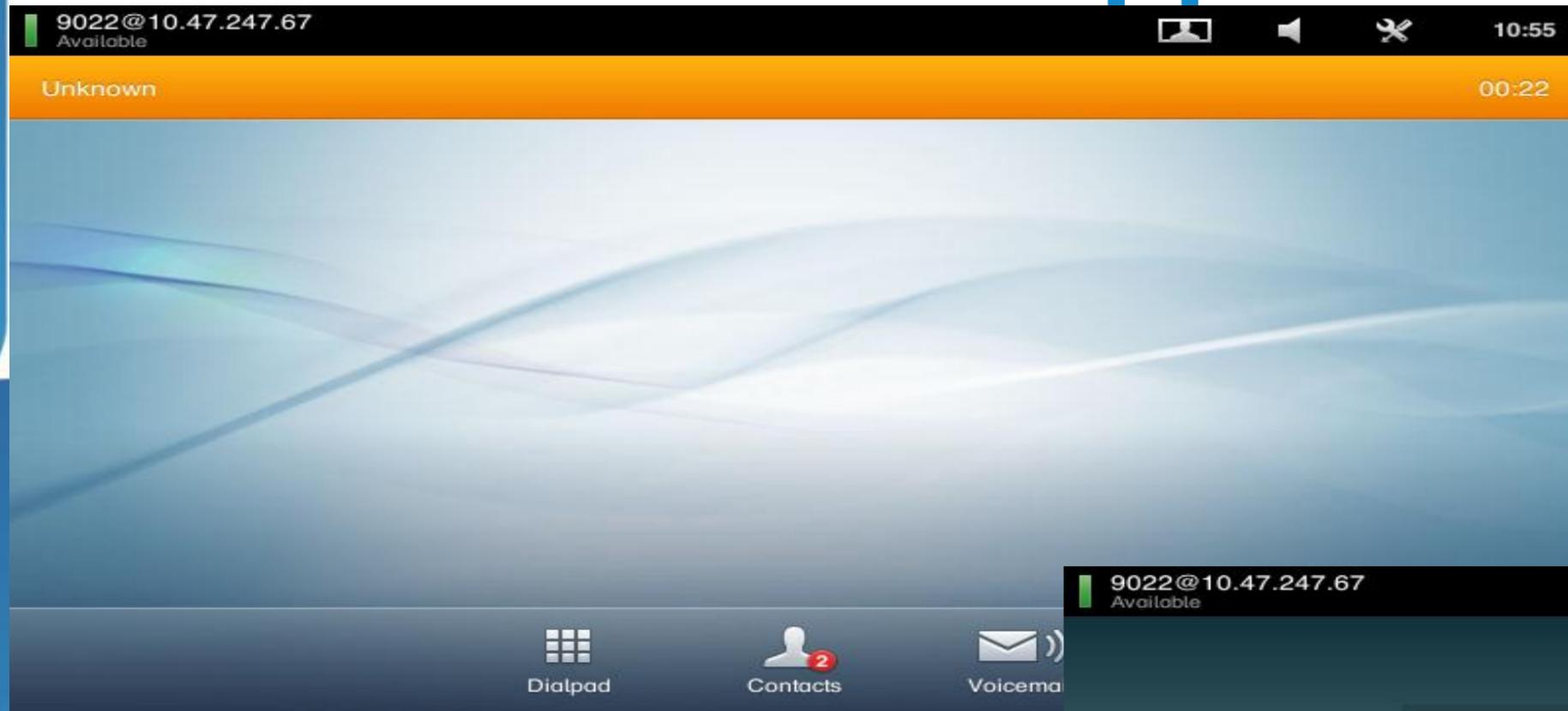
# Voicemail Integration Summary

- All UCM endpoints can be assigned a voicemail profile
- If Unity is connected to UCM, and a VCS endpoint is diverted to voicemail, VCS must be X7.0 or greater
- If Unity is directly integrated with VCS, and a VCS endpoint is diverted to voicemail, VCS must be X6.1 or greater
- VCS endpoints must have a numeric address in order to work properly with voicemail

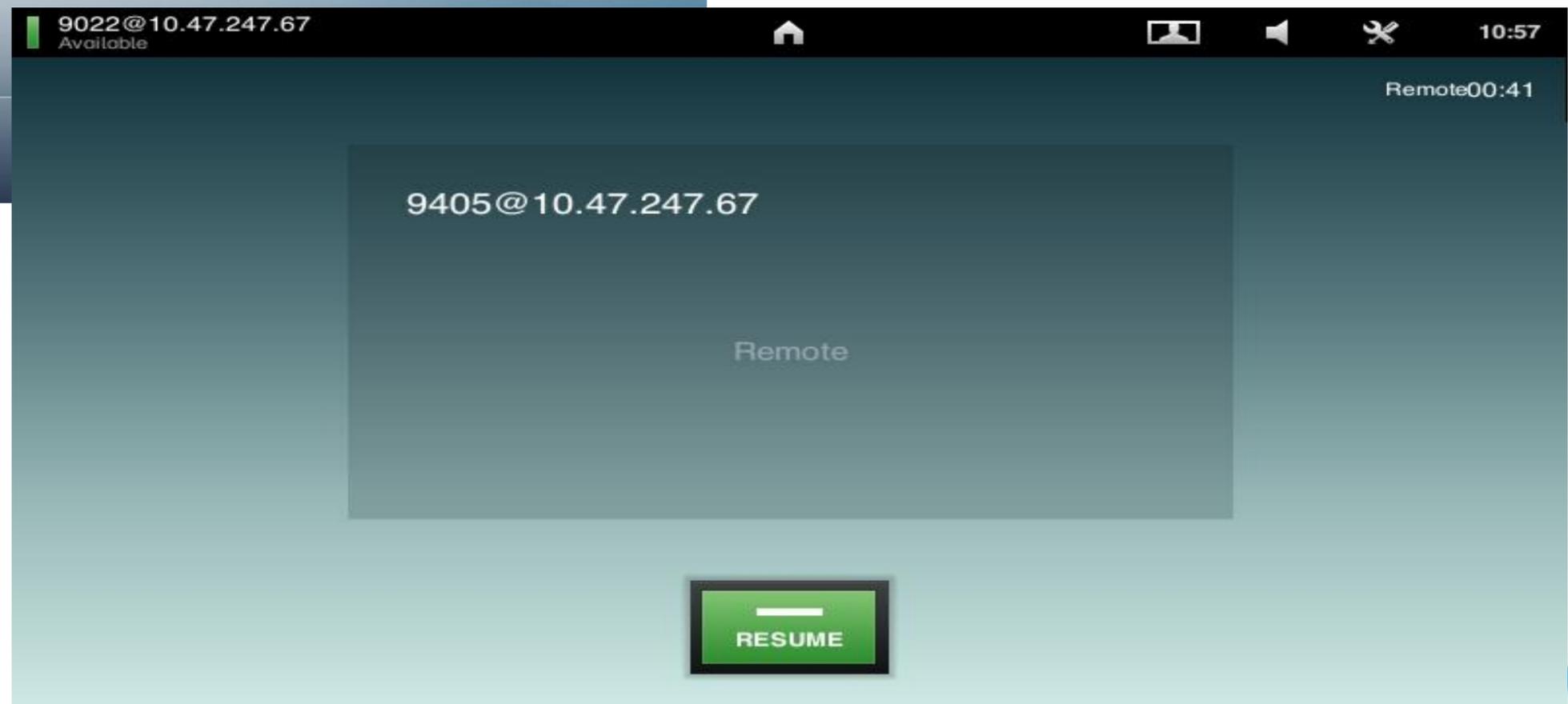
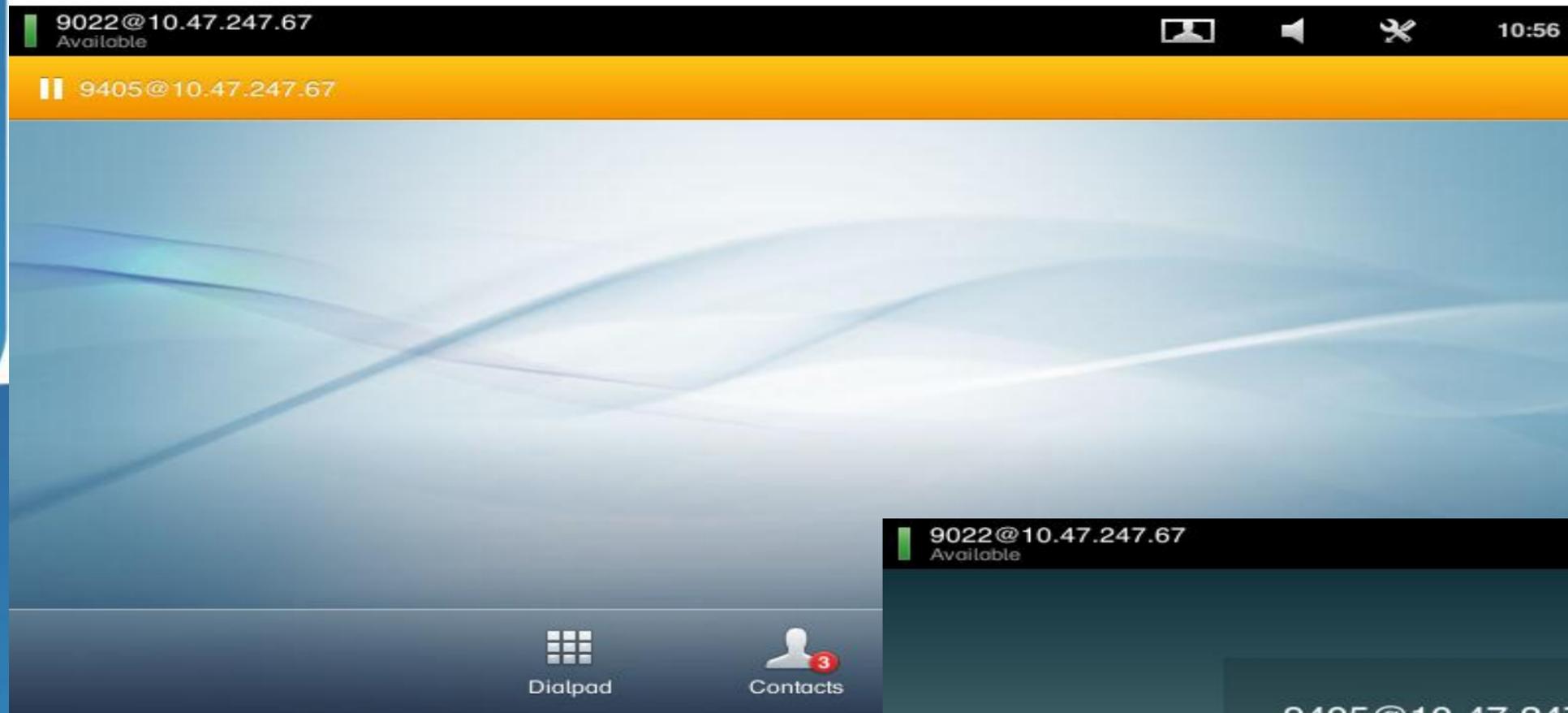
# Shared Lines Support

- EX90 and EX60 now supports shared lines on CUCM.
- Barge call is supported
  - Requires TP MCU resources to support video barge
  - CUCM resources only support audio conferencing
- Supports Resume Remote.

# Shared Lines Support - Barge



# Shared Line Support – Remote Resume



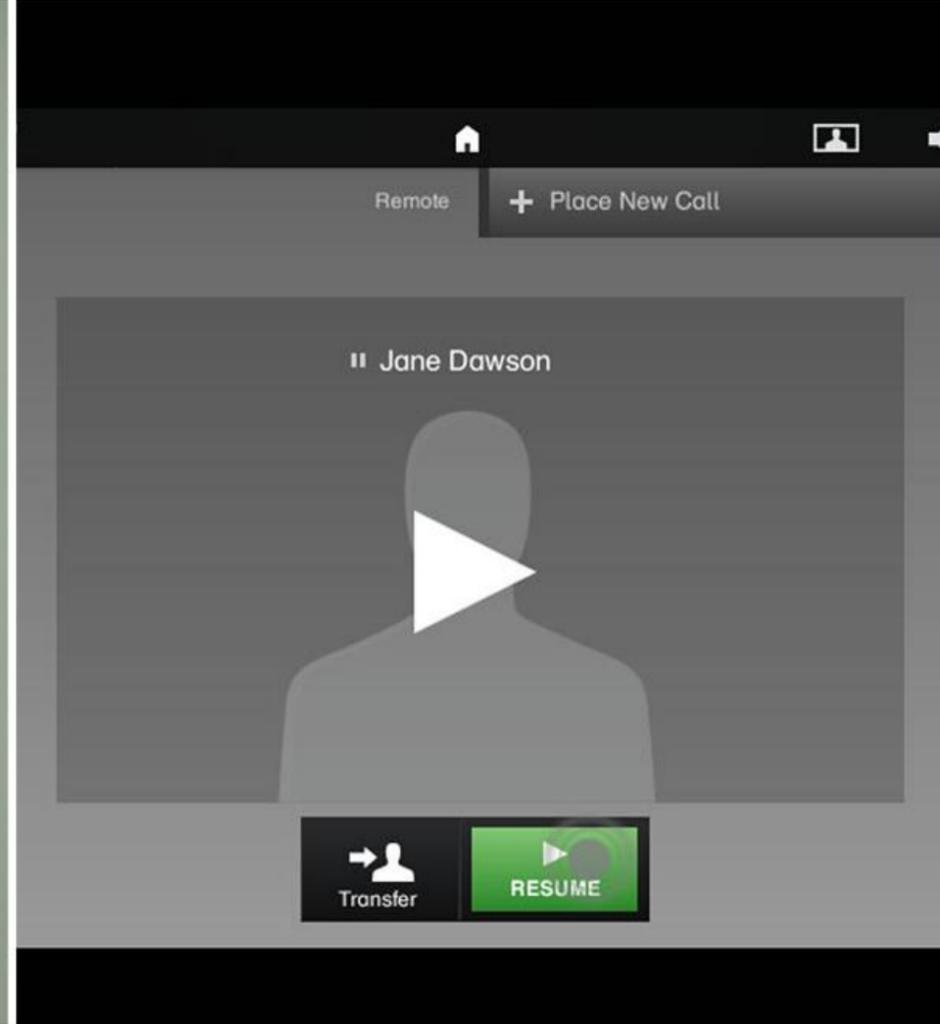
# Shared Lines

- Shared Line on CUCM
  - Same DN/Partition for two or more devices  
(DN not unique, “DN + Partition” forms the unique identifier)
- Shared Line Appearances
  - Remote State Notifications
  - Hold / Resume
  - cBarge
  - Unified Mobility
    - Hand-off from mobile to desk
    - No handoff from desk to mobile

# Seamless mobility with SNR



1. Pick up a call on mobile



2. Resume call on the EX



3. Automatic escalation to video

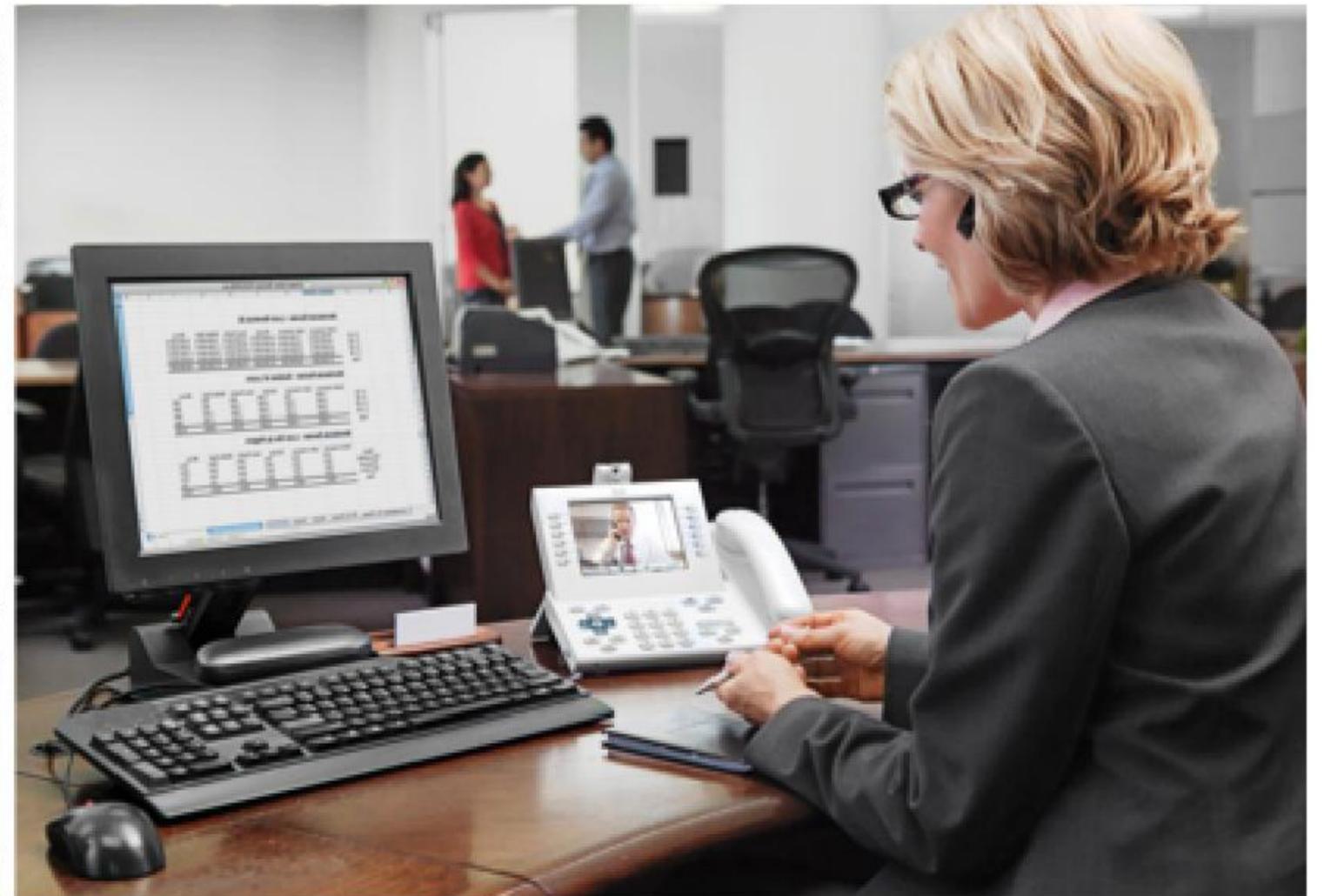
# Shared Line Support



## Assisted Call handling

administrative assistant

Admins can now fully manage calls and lines for executives using EX Systems



# Shared lines - Helpdesk

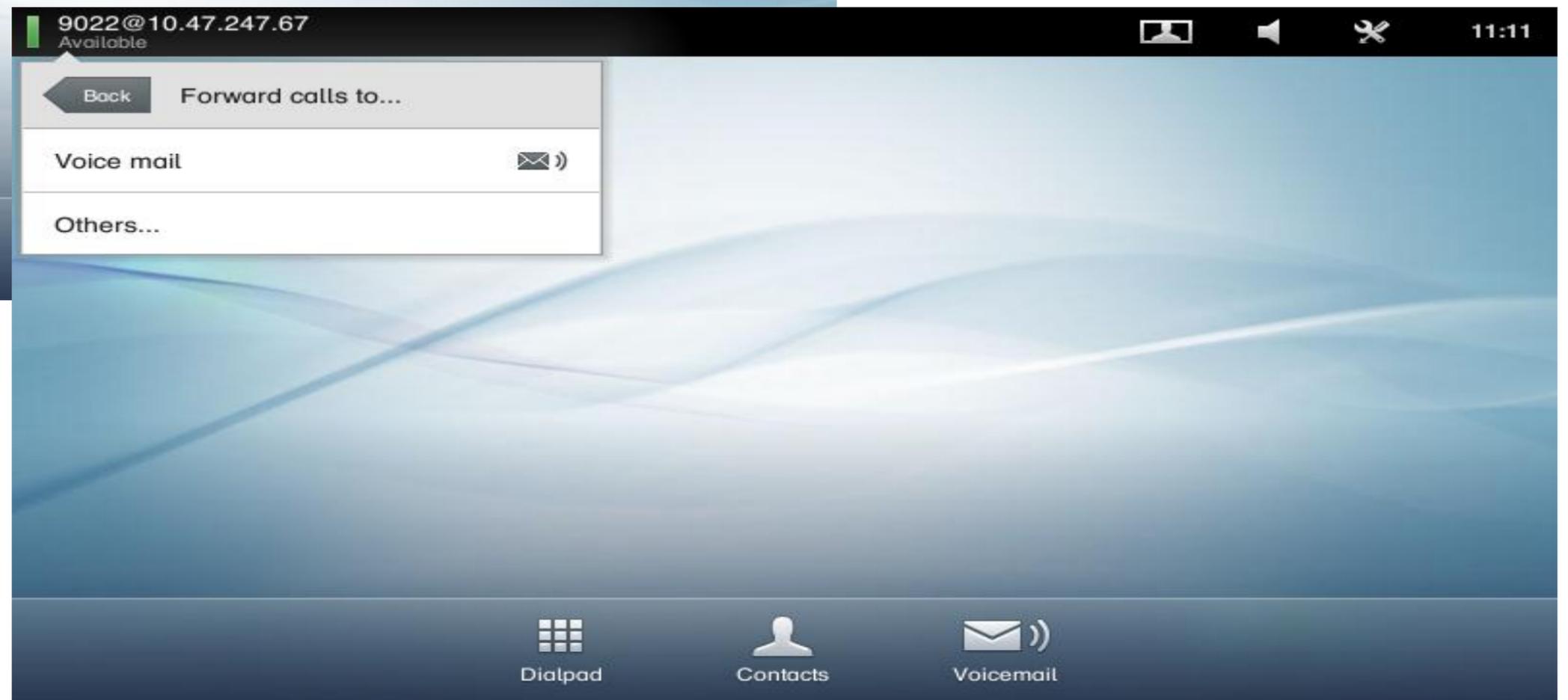
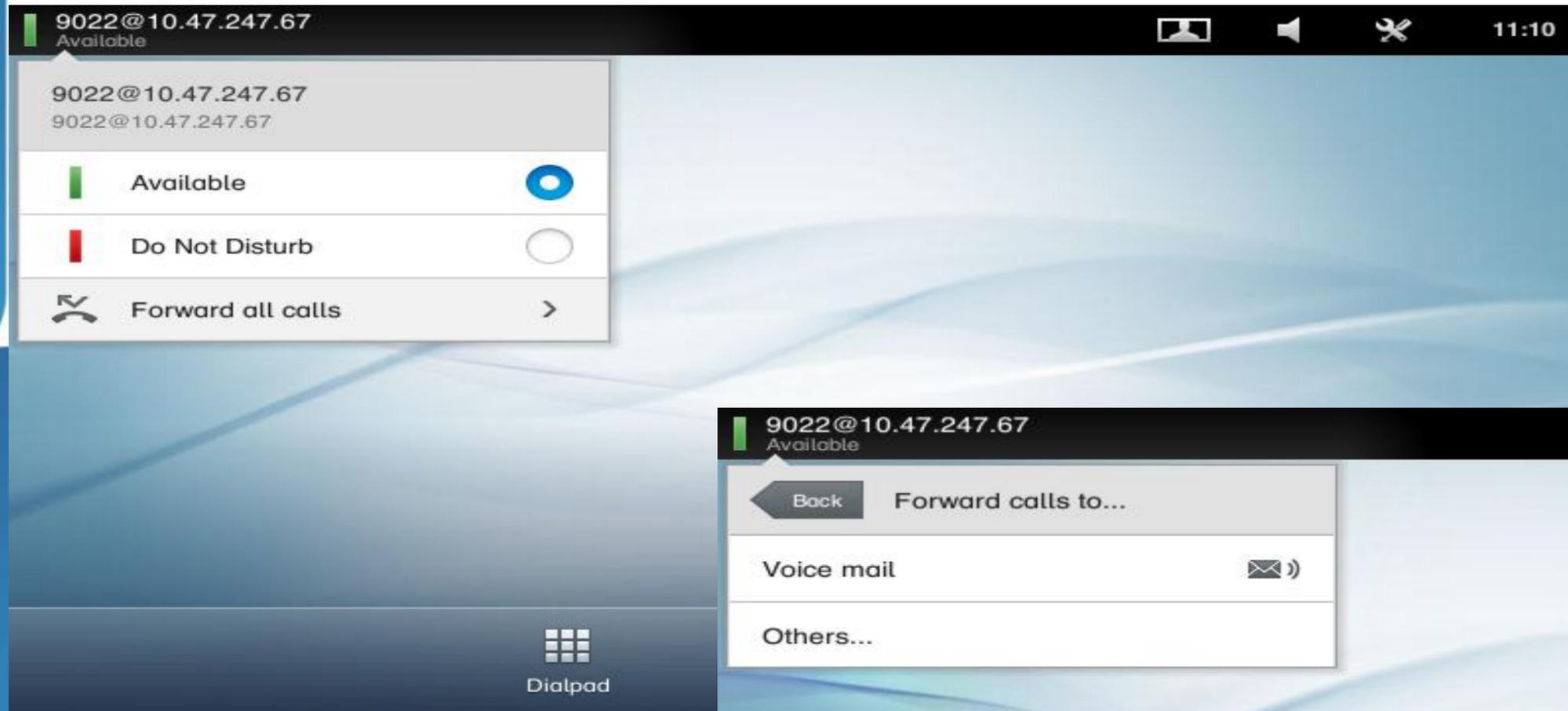


# Consultative Transfer

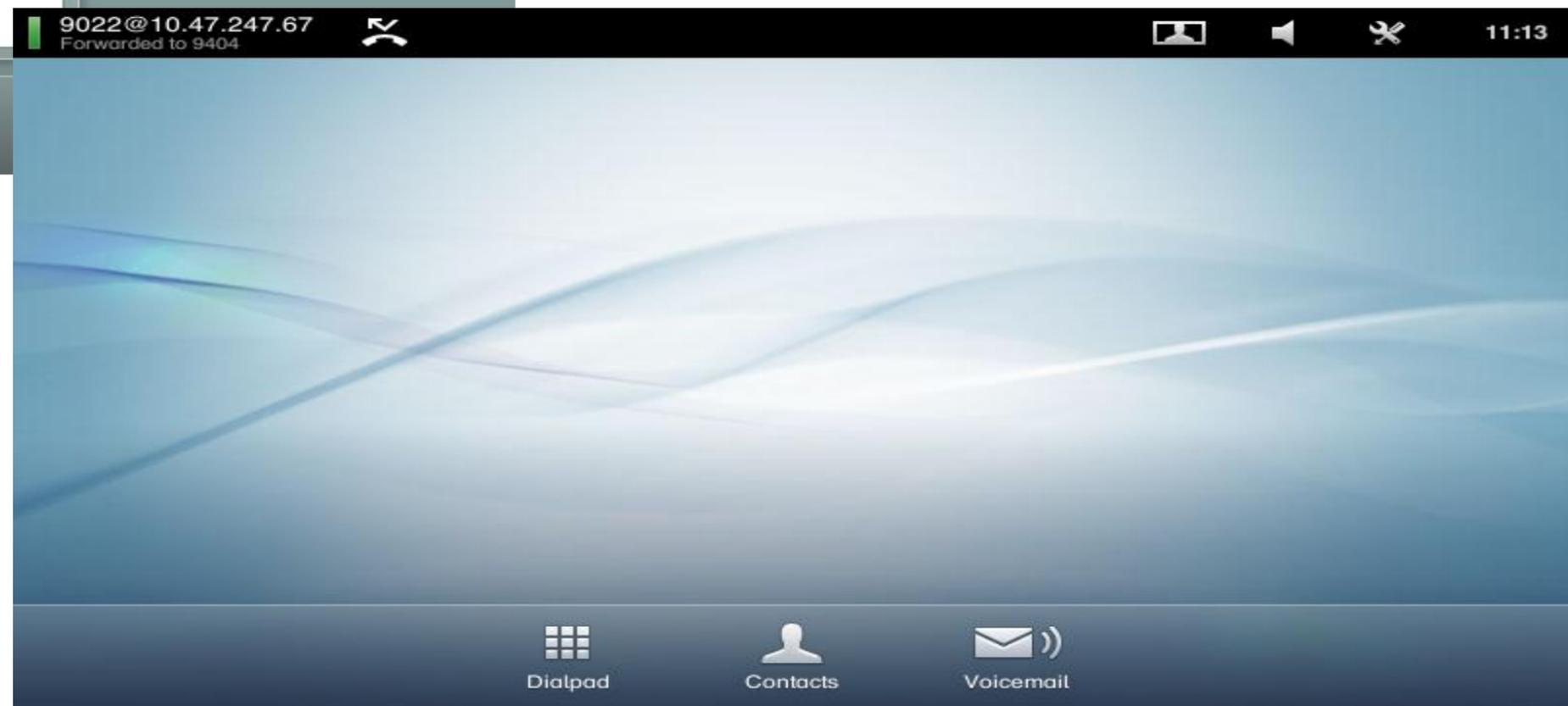
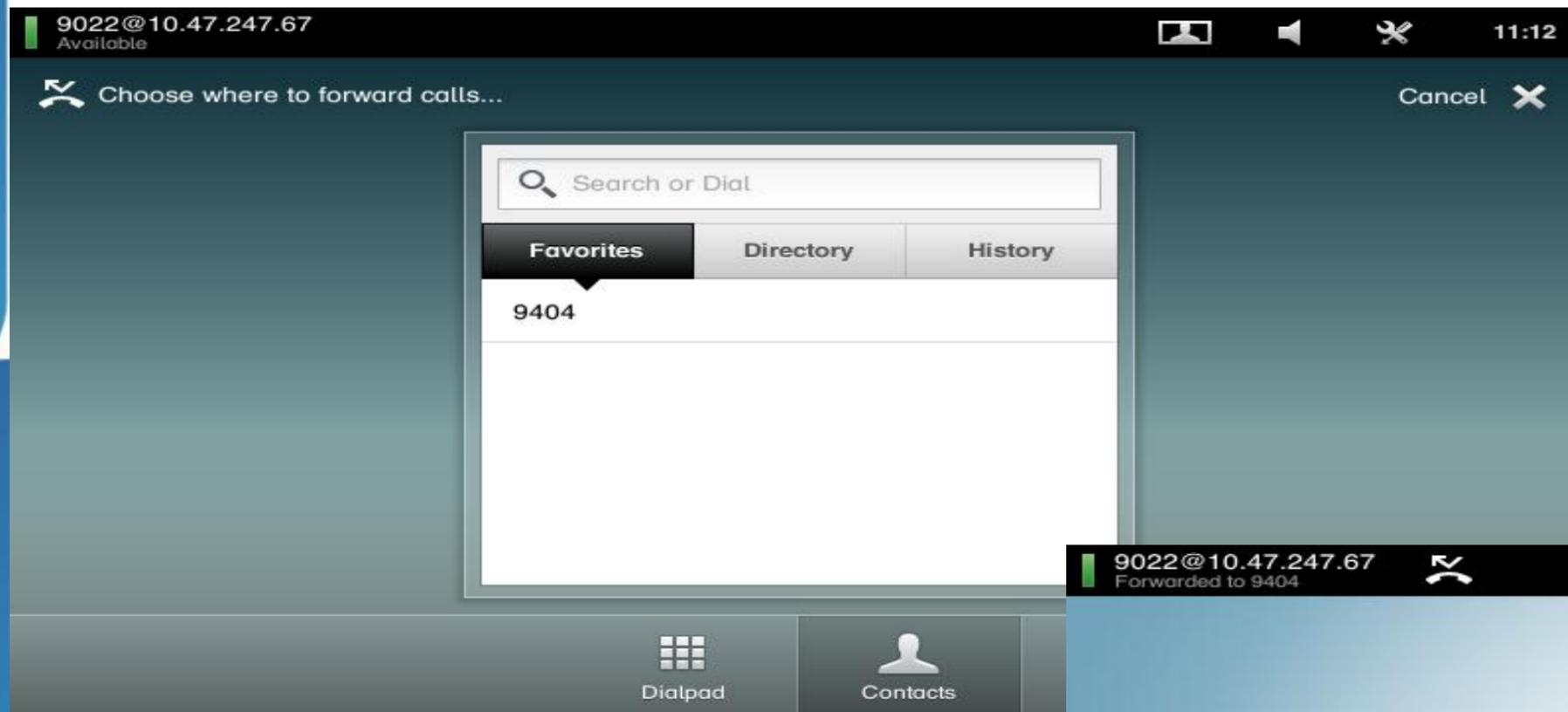


Consultative Transfer Interaction

# Call Forward All Calls



# Call Forward All Calls, enabled



# Other Sessions of Interest

- **Orchestrate and virtualise conferencing resources**
  - BRKEVT-2809
  - Understanding Cisco TelePresence Conductor
- **Use TMS to schedule video endpoints**
  - BRKEVT-2664
  - Video Communications Management and Scheduling
  - Learn how to deploy TelePresence CAC
- **BRKUCC-2667**

Unified CM Enhanced Locations CAC Design Session and Deployment

# Q & A



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