Test Results for Cisco Unified Communications System Release 10.5 Phase -I for Japan

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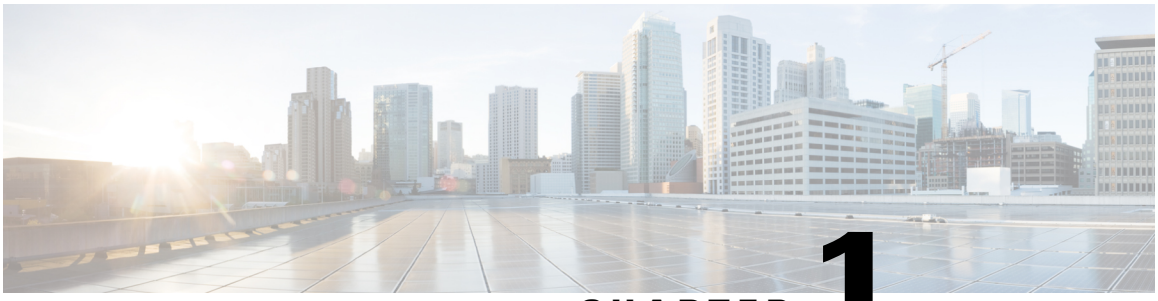

CHAPTER 1

Cisco Unified Communications System Test

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Cisco Unified Communications System Test

Cisco Unified Communications System Test, an integral part of the Enterprise Voice Solution Management is a program that validates and tests specified systems-level solution for the various products and platforms in the Cisco Unified Communications System.

Cisco Unified Communications System Test, the systems integration layer, ensures that the Unified Communications components delivered across the various engineering teams when combined, improves the Unified Communications System software quality. This is achieved by testing the various components.

The requirements for Cisco Unified Communications System Test is derived based on the following:

- Popular customer scenarios
- Input from various Business Units, fields and Cisco Services

The test bed architecture is built based on the Solution Reference Network Design (SRND), cross-section of product deployment models etc. The different types of testing carried out as part of Cisco Unified Communications System Test are:

- Interoperability/Compatibility
- Functionality
- Availability/Reliability/Stability
- Performance/Scalability/Capacity
- Usability/Serviceability
- Special focus area - CAP (Customer Assurance Program)/Technical Assistance Center (TAC)
- Security

Cisco Unified Communications System Test for Japan

Cisco Unified Communications System Test for Japan, in turn is an add-on testing at the solution level, where the requirements gathered are specific to Japanese usage and market. The requirements are derived based on the following:

- Customer found defects in selected UC products
- High priority cases that are covered by the Cisco Unified Communications System Test team
- Inputs from SEs, TAC team of Cisco Japan

The test execution is carried out on selected UC products, which affect the Japanese segment and that are prioritized by SEs of the Cisco Japan team. Japanese specific equivalents, such as, Japanese locale, ISDN Switch type being NTT, JPNP for Numbering Plan are implemented.

The objective of Cisco Unified Communications System Test for Japan is to run a sub-set of system testing that is not covered by Cisco Unified Communications System Test and implement equivalents with Japanese environment such as Japanese OS, localized application, select Cisco Compatible Products, and third party equipment.

In this Cisco Unified Communications System Test release for Japan, the following components are tested.

- Cisco Unified Communications Manager
- Cisco TelePresence Video Communication Server
- Cisco Jabber for iPhone and iPad
- Cisco Jabber for Android
- Cisco Jabber for Windows
- Cisco Jabber for Mac
- Cisco Unified Border Element
- Cisco Unified Survivable Remote Site Telephony
- Cisco Unified Communications Manager IM and Presence
- Cisco Unity Connection
- Cisco Unified Attendant Console
- Cisco UC Integration ™ for Microsoft Lync
- Cisco TelePresence Multipoint Control Unit
- Cisco TelePresence Management Suite
- Cisco TelePresence Conductor
- Cisco TelePresence Server 7010
- Cisco TelePresence Server on VM
- Cisco Collaboration Expressway

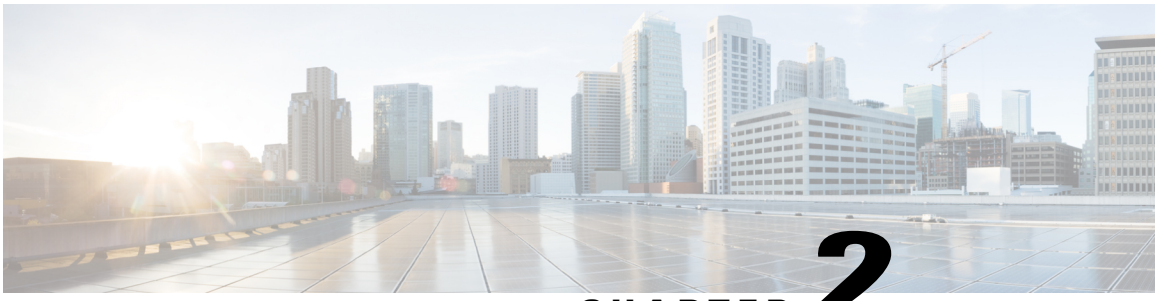
Acronyms

Acronym	Description
AMWI	Audible Message Waiting Indicator
AAR	Automated Alternate Routing
ANAT	Alternate Network Address Translation
ACN	Alternate Contact Number
ACD	Automatic Call Distribution
AGC	Automatic Gain Control
ASA	Adaptive Security Appliance
ASCII	American Standard Code for Information Interchange
ATA	Analog Telephone Adapter
BAT	Bulk Administration Tool
BLF	Busy Lamp Field
CA	Certificate Authority
CAS	Channel Associated Signaling
CCD	Call Control Discovery
CDP	Cisco Discovery Protocol
CDR	Call Detail Record
CED	Caller Entered Digits
CFA	Call Forward All
CFB	Call Forward Busy
CFD	Customer Found Defect
CFNA	Call Forward No Answer
CFNC	Call Forward No Coverage
CFUR	Call Forward Unregistered
CJA	Cisco Jabber for Android
CJI	Cisco Jabber for iPhone
CJIP	Cisco Jabber for iPad
CJW	Cisco Jabber for Windows
CLI	Command Line Interface
CLID	Calling Line Identification

Acronym	Description
CoW	Clustering over WAN
CSF	Client Services Framework
CSS	Calling Search Space
CTI	Computer Telephony Interface
CTI	Computer Telephony Integration
CTL	Certificate Trust List
CUAC	Cisco Unified Attendant Console
CUBE	Cisco Unified Border Element
CUC	Cisco Unity Connection
CUCI-Lync	Cisco UC Integration ™ for Microsoft Lync
CUCM	Cisco Unified Communications Manager
DCR	Device and Credential Repository
DHCP	Dynamic Host Configuration Protocol
DID	Direct In-Ward Dialing
DN	Directory Number
DND	Do Not Disturb
DO	Delayed Offer
DPNSS	Digital Private Network Signaling System
DSCP	Differentiated Services Code Point
EM	Extension Mobility
EMCC	Extension Mobility Cross Cluster
EO	Early Offer
FXS	Foreign Exchange Station
GW	Gateway
GUI	Graphical User Interface
HR	Historical Reporting
HA	High Availability
HD	High Definition
HTML	Hyper Text Markup Language
ICT	Inter Cluster Trunk
IM	Instant Messaging

Acronym	Description
IPPM	IP Phone Messenger
ISDN	Integrated Services Digital Network
ITL	Initial Trust List
KEM	Key Expansion Module
LDAP	Lightweight Directory Access Protocol
MCU	Multipoint Control Unit
MGCP	Media Gateway Control Protocol
MOH	Music On Hold
MRGL	Media Resource Group List
MWI	Message Waiting Indicator
NLP	Non Linear Processing
NTLMv2	New Technology LAN Manager version 2
NTP	Network Time Protocol
OSD	On Screen Display
POTS	Plain Old Telephony System
PCA	Personal Communication Assistant
PCoIP	PC over IP
PFS	Peer Firmware Sharing
PIN	Personal Identification Number
PiP	Picture in Picture
PRI	Primary Rate Interface
PRT	Problem Reporting Tool
PSTN	Public Switched Telephone Network
QRT	Quality Report Tool
QSIG	Q-Signaling protocol
RSS	Really Simple Syndication
RTMT	Real Time Monitoring Tool
RTP	Realtime Transport Protocol
SAML	Security Assertion Markup Language
SIP	Session Initiation Protocol
SCCP	Skinny Client Control Protocol

Acronym	Description
SD	Standard Definition
SRST	Cisco Unified Survivable Remote Site Telephony
SSH	Secure Shell
SSL	Secure Socket Layer
SSO	Single Sign On
TAC	Technical Assistant Center
TCP	Transmission Control Protocol
TLS	Transport Layer Security
TRP	Trust Relay Point
TUI	Telephony User Interface
UCS	Unified Computing System
UDS	User Data Service
UMG	Unified Messaging Gateway
URI	Uniform Resource Identifier
VCS	Cisco TelePresence Video Communication Server
VGW	Voice Gateway
VoIP	Voice over IP
VPIM	Voice Profile for Instant Messaging
VMN	Voice Mail Notification
VMO	View Mail for Outlook
VPN	Virtual Private Network
VTS	TelePresence Server on VM
WAN	Wide Area Network
Wi-Fi	Wireless Fidelity
WLC	Wireless LAN Controller



CHAPTER 2

Test Topology and Environment Matrix

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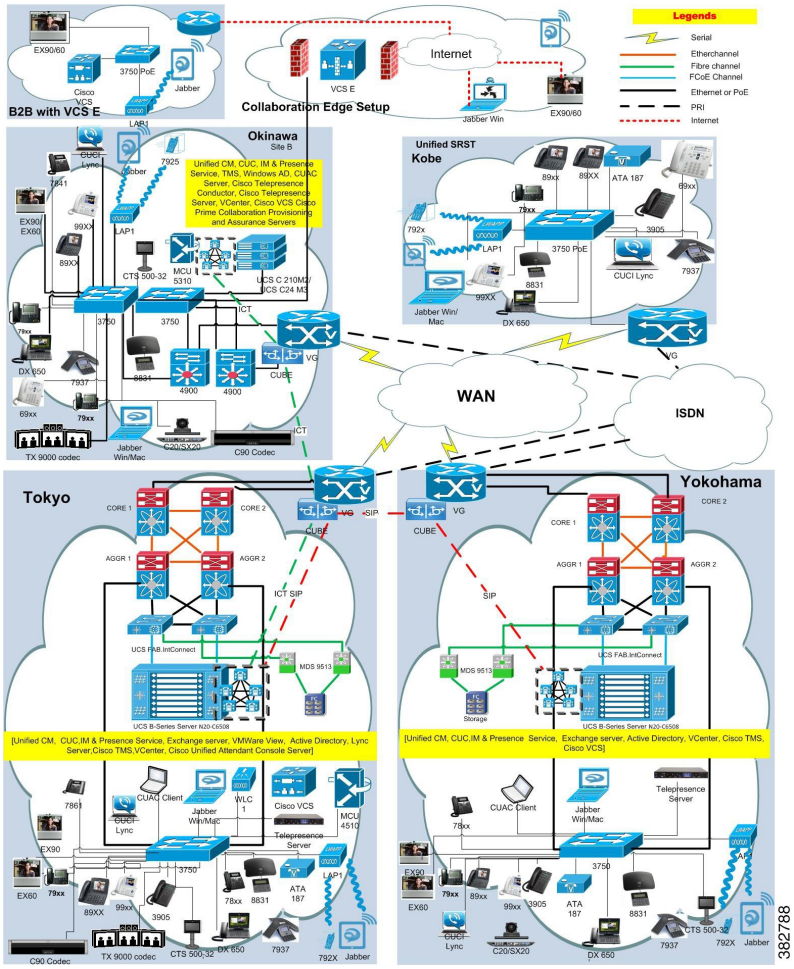
Test Topology

Figure 1: Topology in Use



Environment Matrix

Applications	Component		Version
Call Control	Cisco Unified Communications Manager	Version	10.5.1.10000-7
		Locale	10.5.1.9903-101
		Dial Plan	3-1-9.JP
	Cisco Unified Survivable Remote Site Telephony (SRST)	Version	10.0(1)
		IOS	15.4.2.T
	Cisco Telepresence Video Communication Server Expressway (VCS Expressway)	Version	X8.1.1
		Locale	X7.2_LanguagePacks_BETA
	Cisco Telepresence Video Communication Server (VCS)	Version	X8.1.1
		Locale	X7.2_LanguagePacks_BETA
Applications	Cisco Unified Attendant Console	Version	10.5.1.5-98
	Cisco Unified Communications Manager IM and Presence Service	Version	10.5.1.10000-9
		Locale	10.5.1.1000-1
Voice Mail and Unified Messaging	Cisco Unity Connection	Version	10.5.1.10000-7
		Locale	10.5.0.1-1
	View Mail for Outlook	Version	10.0(1)
	Visual Voicemail	Version	8.6

Applications	Component		Version
End Point	Cisco Unified IP Phones 6921/41/45/61		9.4.1.3
	Cisco IP Phones 7821/41/61		10.1.1.SR1-4
	Cisco Unified IP Phones 7942/62/75		9.3-1 SR4-1S
	Cisco Unified Wireless IP Phones 792X		1.4.5SR1.3
	Cisco Unified IP Conference Phone 8831		9-3-3-5
	Cisco Unified IP Phones 8941/8945		9.4.1.SR1.1
	Cisco Unified IP Phones 9951/9971		9.4.1. 9
	Cisco Desktop Collaboration Experience DX650		10.2.1JBT0-440
	Cisco ATA 187 Analog Telephone Adaptor		9.2.3.1
	Cisco UC Integration™for Lync		9.6.0.621
	EX60 - Cisco TelePresence System EX60		TC 7.1.2
	EX90 - Cisco TelePresence System EX90		TC 7.1.2
	SX20 - Cisco TelePresence SX20 Quick Set		TC 7.1.2
	Cisco TelePresence System Integrator Package C90		TC 7.1.2
	CTS 500 - Cisco Tele Presence System 500 (32)		TX6.1.2.1(5)
	Cisco TX6.1.2.1(5) TelePresence TX9000		TX6.1.2.1(5)

Applications	Component		Version
Communications Infrastructure	Gateways	IOS	15.4(2) T1
	Cisco Unified Border Element for ISR		10.0.0
	Cisco 3750 PoE Switch		15.0.2-SE 5
	VCenter Server		ESXi 5.1.0
	MDS Switch	M9500	5.2(2 a)
TelePresence	Cisco Telepresence Management Suite-TMS	Version	14.4
	Cisco TelePresence Management Suite Provisioning Extension-> TMSPE	Version	1.2
	MCU 4510 & 5310 - Cisco Telepresence MCU	Version	4.4(3.67)
		Locale	MCU_4-3_UI_and_audio_JPN.package
	Cisco Telepresence Server on VM	Version	4.0(1.57)
	Cisco Telepresence Conductor	Version	XC2.3
	Cisco Telepresence Server 7010	Version	4.0(1.57)
Wireless and Mobility	Wireless LAN Controller 44XX		7.0.250.0
	Wireless Access Point 1142		15.2
	Wireless Access Point 35XX		15.2
	Cisco Jabber for Mac		9.6.0.173114
	Cisco Jabber for Windows		10.5.0.28285
	Cisco Jabber for iPhone and iPad	Version	10.5.0.178947
		iPhone 5	Apple iOS 7.0.4
		iPad	Apple iOS 7.0.4
	Cisco Jabber for Android	Version	10.5.0.178897
		Galaxy SII	Android OS 4.0.3
		Galaxy S4	Android OS 4.3

Applications	Component		Version
UCS	Fabric Interconnect PRIMARY	Cisco UCS 6140	2.1.2a
	Fabric Interconnect SUBORDINATE	Cisco UCS 6140	2.1.2a
	Fabric Cluster	Cisco UCS 6140	2.1.2a
	ESXi Host	Blade Server-1	ESXi 5.1.0
Client	Operating System	Windows 7 -SP1	Windows 7 - SP1 (Japanese)
		Windows 8.1	Windows 8.1 (Japanese)
		Mac	10.8
	Browser	IE	IE 11 (Supported Japanese language)
		Mozilla	Firefox 29.0.1 (Supported Japanese language)
		Chrome	Chrome 35 (Supported Japanese language)
	Microsoft Lync Client		2010
	Microsoft Outlook		2010
Server	Windows Server		Windows Server 2008(R2 Enterprise - Japanese)
			Windows Server 2012 (Japanese)
	Exchange Server		2010
	Lync Server		2010

Note Install Cisco Unified Communications Manager Locale in Unity Connection to support Linguistic command strings of "SAML Single Sign On" pages and Login page of Unity Connection. The "SAML Single Sign On" is basically the Cisco Unified Communications Manager page being displayed in Unity Connection Administration .

Procedure to Install Unified CM Locale in Unity Connection :

- Download required Unified CM locale version from CCO

Note The Version Details are given in the above Environment Matrix Table

- Sign in to Cisco Unity Connection Serviceability

- Stop the Connection Conversation Manager and Connection Mixer services
- Sign in to Cisco Unified Operating System Administration and select Software Upgrades menu, select Install/Upgrade. The Software Installation/Upgrade window appears.
- In the Source list, select Remote File system
- In the Directory field, enter the path to the folder that contains the language file on the remote system
- In the Server field, enter the FTP / SFTP server name or IP address.
- In the User Name field, enter your user name on the remote server.
- In the User Password field, enter your password on the remote server.
- In the Transfer Protocol list , select FTP /SFTP option.
- Select Next.
- Start the Connection Conversation Manager and Connection Mixer services

*If you installed Japanese: Run the CLI command **utils service restart Cisco Tomcat** to restart the Tomcat service.*

Open Caveats

Defect ID	Title
Cisco Unified Communications Manager	
CSCuo80008	Different Call Details in SIP and SCCP phones of 894x
Cisco VCS	
CSCuo78101	Forwarded Call Information Display configuration is not reflected
CSCuo88112	Hold icon near participant's DN is not displayed in a conference call
CSCuo90595	Unable to transfer call from EX90 during scheduled conference
CSCuo99704	Wrong participant list showing in TX9000 after a directed call park
CSCuo87291	Directed call park showing as conference in EX90
CSCuo47595	Wrong DN displayed in EX90

What's New?

Single Sign On

The Single Sign On feature allows end users to log into a Windows client machine on a Windows domain, and then use certain Cisco Unified Communications Manager applications without signing on again.

Single sign-on (SSO) is an authentication process that allows a user to access multiple applications with one set of login credentials

CHAPTER 3

Test Results Summary

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Cisco Unified Communications Manager

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCMG120	Create SIP Profile with early offer support as Disabled and make an inter-cluster call from SIP end point	Verify that the inter cluster call via SIP Trunk with the early offer as disabled in the SIP Profile is successful	IP Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> IP Phone B	Passed	
UCJ10SCUCMG121	Create SIP Profile with early offer support as Best Effort and make an inter-cluster call from SIP end point	Verify that the inter cluster call via SIP Trunk with the early offer as Best Effort in the SIP Profile is successful	IP Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> IP Phone B	Passed	
UCJ10SCUCMG126	Create SIP Profile with early offer support as Mandatory and make an inter-cluster call from SCCP end point	Verify that the inter cluster call via SIP Trunk with the early offer as Mandatory in the SIP Profile is successful	IP Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> IP Phone B	Passed	
UCJ10SCUCMG137	Make a conference call from 88xx IP Phone to other IP Phones	Verify whether conference call can be made from 88xx IP Phone to other IP Phones successfully	88xx IP Phone A -> Unified CM -> IP Phone A; 88xx IP Phone A -> Unified CM -> IP Phone B	Passed	
UCJ10SCUCMG138	Make a conference call from 88xx IP Phone to other IP Phones via SIP Trunk	Verify whether conference call can be made from 88xx IP Phone to other IP Phones via SIP trunk successfully	88xx IP Phone A -> Unified CM1 -> IP Phone A; 88xx IP Phone A -> Unified CM1 -> SIP trunk -> Unified CM2 -> IP Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCMG159	Hold and Resume shared line call in 88XX IP phone	Verify whether call can be put on hold and resume in 88XX IP Phone when in shared line.	IP Phone A -> Unified CM -> 88XX IP Phone A -> Unified CM -> 88XX IP phone B	Passed	
UCJ10SCUCMG026	SIP URI dialing from Jabber Client1 to Jabber Client2 intra cluster	Verify that the intra cluster SIP URI dialing between two Jabber clients is successful	Jabber Client1 -> Unified CM -> Jabber Client2	Passed	
UCJ10SCUCMG082	SIP URI dialing from Jabber Client1 to IP Phone A (99xx) and Call Pickup at IP Phone B at the same cluster	Verify that the intra cluster SIP URI dialing between Jabber client and IP Phone is successful and call pickup is done at the other IP Phone	Jabber Client1 -> Unified CM -> IP Phone A (99xx) -> Unified CM -> IP Phone B	Passed	
UCJ10SCUCMG083	SIP URI dialing from Jabber Client1 to IP Phone A (99xx) and Call forward all from IP Phone A to IP Phone B at the same cluster	Verify that the intra cluster SIP URI dialing from Jabber Client1 to IP Phone A (99xx) and Call forward all from IP Phone A to IP Phone B at the same cluster is successful	Jabber Client1 -> Unified CM -> IP Phone A (99xx) -> Unified CM -> IP Phone B	Passed	
UCJ10SCUCMG124	Make a call with MGCP Gateway registered analog phone using NIM-4FXO interface	Verify that the call made with MGCP Gateway registered analog phone using NIM-4FXO interface is successful	Analog Phone -> Unified CM -> IP Phone A	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCMG125	Register MGCP Gateway in Unified CM using NIM-2FXS/4FXO interface	Verify that the registration of MGCP Gateway in Unified CM with Cisco4451 ISR using NIM-2FXS/4FXO interface is successful	NA	Passed	
UCJ10SCUCMG128	Make a call with SCCP Gateway registered analog phone using NIM-2FXS-SCCP interface	Verify that the call made with SCCP Gateway registered analog phone using NIM-2FXS-SCCP interface is successful	Analog Phone -> Unified CM -> IP Phone A	Passed	
UCJ10SCUCMG130	Make a call with SCCP Gateway registered analog phone using NIM-4FXS-SCCP interface	Verify that the call made with SCCP Gateway registered analog phone using NIM-4FXS-SCCP interface is successful	Analog Phone -> Unified CM -> IP Phone A	Passed	
UCJ10SCUCMG133	Generate Multi-server CSR on any server and upload the corresponding Certificate from single server in the cluster	Verify that Multi-server CSR generation and upload into the cluster is successful	NA	Passed	
UCJ10SCUCMG005	Launching TMS via Self-Care portal by setting the Enterprise parameter as "Show as Browser popup"	Verify that the TMS is launched successfully via Self-Care portal by setting the Enterprise parameter as "Show as Browser popup"	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCMG016	Auto login when user launches the Unified CM Administration web page	Verify that the Unified CM Admin page is logged in automatically without prompting any user credentials	NA	Passed	
UCJ10SCUCMG303	Default call protocol changed to SIP in Cisco Telepresence System EX series Phone page	Verify whether the Default call protocol set to SIP in EX-series Phone page is updated in Unified CM Phone Page for the particular Phone	NA	Passed	
UCJ10SCUCMG305	Multi point mode is changed to CUCM MRGL in Cisco Telepresence system EX series Phone page	Verify whether the Multi point mode set to CUCM MRGL in EX-series Phone page is updated in Unified CM Phone Page for the particular Phone	NA	Passed	
UCJ10SCUCMG311	OSD encryption is enabled in Cisco Telepresence system EX series Phone page	Verify whether the OSD encryption enabled in EX-series Phone page is updated in Unified CM Phone Page for the particular Phone	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCMG327	Microsoft AD 2012 R2 integration with Unified CM 10.5	Verify whether the AD 2012 R2 is integrated with Unified CM 10.5 by creating an user in AD and syncing with Unified CM	NA	Passed	
UCJ10SCUCMG201	Pages of Unified CM on Browser by clicking on the links are Navigating to Corresponding Pages on IE10/11	Verify that the Pages of Unified CM on Browser by clicking on the links are navigating to corresponding pages and displaying clearly on IE10/11	NA	Passed	
UCJ10SCUCMG204	Pages of CUCM on Browser by clicking on the links Navigating to Corresponding Pages on Latest Version of Firefox	Verify that Pages of CUCM on Browser by clicking on the links are navigating to corresponding pages and displaying clearly on Latest Version of Firefox	NA	Passed	
UCJ10SCUCMG021	Auto login when user launches the Cisco Self-Care portal	Verify that the Cisco Self-Care portal is logged in automatically without prompting any user credentials	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCMG022	Auto login to Unified CM and phone registration for making calls between two IP phones	Verify whether the Admin page of Unified CM is launched automatically without prompting any user credentials and phone registration is also successful for making calls between two IP phones	Phone A -> Unified CM -> Phone B	Passed	
UCJ10SCUCMG023	Auto login into Cisco Self-Care portal to enable DND for the IP Phone associated with the end user	Verify whether the user can login into Self-Care portal automatically without prompting for any user credentials and can able to enable the DND for the IP Phone associated with the end user	NA	Passed	
UCJ10SCUCMG024	Auto login into Cisco Self-Care portal and assigning the speed dial to the phones associated with the end user via Self-Care	Verify whether the user can login into Self-Care portal automatically without prompting for any user credentials and assigning the phones with speed dial to call the other IP phone	Phone A -> Unified CM -> Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCMG407	All nodes in the cluster are auto populated while generating Multi-server (SAN) certificate	Verify whether all the nodes in the cluster are auto populated when issuing the Multi-server (SAN) certificate	NA	Passed	
UCJ10SCUCMG103	Create SIP Profile with early offer support as Mandatory and make an inter-cluster call from SIP end point	Verify that the inter cluster call via SIP Trunk with the early offer as Mandatory in the SIP Profile is successful	Phone A -> Unified CM1 ->SIP Trunk -> Unified CM2 -> Phone B	Passed	
UCJ10SCUCMG402	Call details on 894x phones after call made to the Hunt Pilot Number	Verify whether the call details of 894x phones are proper when call made to hunt pilot number	Phone A -> Unified CM -> Hunt Pilot -> Unified CM -> Phone B	Failed	CSCuo80008

Cisco TelePresence Video Communication Server

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG001	Modifying MTU Size under System->IP page in Cisco VCS	To verify that user is able to change the MTU size under System->IP page in Cisco VCS	NA	Passed	
UCJ10SSVCSG002	Modifying MTU Size via CLI in Cisco VCS	To verify that user is able to change the MTU size via CLI in Cisco VCS	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG003	Modifying the MTU Size and making a call between Cisco VCS registered endpoints	To verify that user is able to change the MTU size and can make a call between Cisco VCS registered endpoints	EX60->Cisco VCS->EX90	Passed	
UCJ10SSVCSG004	Modifying the MTU Size under System->IP for LAN 1 in Cisco VCS Expressway	To verify that user is able to change the MTU size uder System->IP for LAN 1 in Cisco VCS Expressway	NA	Passed	
UCJ10SSVCSG005	Modifying the MTU Size under System->IP for LAN 1 and LAN 2 in Cisco VCS Expressway	To verify that user is able to change the MTU size under System->IP for LAN 1 and LAN 2 in Cisco VCS Expressway	NA	Passed	
UCJ10SSVCSG006	Making a call between Cisco VCS registered endpoints via Expressway after modifying MTU Size in Expressway	To verify that user is able to change the MTU size and make a call between VCS registered endpoints via Cisco VCS Expressway	EX60->Cisco VCS-A->Cisco VCS Expressway->Cisco VCS-B->EX90	Passed	
UCJ10SSVCSG007	Changing SX20 Quick Set Product specific configuration value of telnet mode from on to off in Unified CM	To verify if product specific configuration parameter value of telnet mode is changed from on to off in Cisco TelePresence SX20 Quick Set	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG008	Changing SX20 Quick Set Product specific configuration value of SSH mode from on to off in Unified CM	To verify if product specific configuration parameter value of SSH mode is changed from on to off in Cisco TelePresence SX20 Quick Set	NA	Passed	
UCJ10SSVCSG009	SX20 Quick Set Product specific configuration value of default call protocol set to SIP in Unified CM	To verify if product specific configuration parameter value of default call protocol is set to SIP in Cisco TelePresence SX20 Quick Set	NA	Passed	
UCJ10SSVCSG010	Changing SX20 Quick Set Product specific configuration value of Multipoint mode Set to CUCMMRGL in Unified CM	To verify if product specific configuration parameter value of Multipoint mode is set to CUCMMRGL in Cisco TelePresence SX20 Quick Set	NA	Passed	
UCJ10SSVCSG011	SX20 Quick Set Product specific configuration value of Microphone Unmute On Disconnect Set to On in Unified CM	To verify if product specific configuration parameter value of Microphone Unmute On disconnect Set to On in Cisco TelePresence SX20 Quick Set	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SVCSG012	SX20 Quick Set Product specific configuration value of OSD Encryption Indicator Set to On in Unified CM	To verify if product specific configuration parameter value of OSD Encryption Indicator Set to On in Cisco TelePresence SX20 Quick Set	NA	Passed	
UCJ105SVCSG013	Creating remote support user via Web GUI of SX20 Quick Set	To verify if remote support user is created via Web GUI of Cisco TelePresence SX20 Quick Set registered with Cisco Unified Communications Manager	NA	Passed	
UCJ105SVCSG014	Deleting remote support user via xAPI of SX20 Quick Set	To verify if remote support user is deleted via xAPI of Cisco TelePresence SX20 Quick Set registered with Cisco Unified Communications Manager	NA	Passed	
UCJ105SVCSG015	Vlan mode is set to auto by default in Touch UI of SX20 Quick Set	To verify if Vlan mode is set to auto by default in Touch-UI of Cisco TelePresence SX20 Quick Set	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG016	Vlan mode is set to auto by default in xAPI of EX60	To verify if Vlan mode is set to auto by default in xAPI of Cisco TelePresence System EX60	NA	Passed	
UCJ10SSVCSG017	Root access is disabled in EX60	Verify whether root access is disabled in Cisco TelePresence System EX60 by logging with root credentials in xAPI of EX60	NA	Passed	
UCJ10SSVCSG018	Logging in EX60 using remote support user and password	To verify if it is able to login using remote support user and password in Cisco TelePresence System EX60	NA	Passed	
UCJ10SSVCSG019	Vlan mode is set to auto by default in touch UI of EX60	To verify if Vlan mode is set to auto by default in touch UI of Cisco TelePresence System EX60	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG020	Auto answer in EX60 during call forward	Verify whether call is auto answered in Cisco TelePresence System EX60 registered with Cisco Unified Communications Manager when call is forwarded to Cisco TelePresence System EX60 from Cisco Unified IP Phone 7841 registered with Cisco Unified Communications Manager	6941 -> Unified CM -> 7841 -> Call forward -> EX60	Passed	
UCJ10SSVCSG021	Conference call consultative transfer to EX60	Verify whether conference call consultative transfer from Cisco Unified IP Phone 8941 which is in conference with Cisco TelePresence System EX90 and 6941 registered in Cisco Unified Communications Manager to EX60 registered in Cisco Unified Communications Manager works fine	8941(1) -> Unified CM -> EX90 8941(1) -> Unified CM->6941 -> Conference -> transfer -> Unified CM ->EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG022	Hold and Resume a Shared line call	To verify that user is able to hold the call in 7841 and resume the call from Cisco TelePresence System EX60 while EX60 and 7841 are in Shared line which is registered with Cisco Unified Communications Manager	6921 -> Unified CM -> 7841 and EX60 (Shared line) 7841-> hold -> Unified CM-> EX60 (Resume)	Passed	
UCJ10SSVCSG023	Consultative call transfer from EX60 which is in shared line with EX90	To verify whether Consultative call transfer from EX60(1) which is in shared line with EX90 to EX60(2) registered with Unified CM works successfully	7841 -> Unified CM -> EX60(1) EX90 -> Hold/Transfer-> Unified CM -> EX60(2)	Passed	
UCJ10SSVCSG024	Blind call transfer in EX60 during presentation sharing	To verify whether EX60(1) which is sharing presentation with EX60(2) can do a blind transfer to EX90 registered with Cisco Unified Communications Manager	EX60(1) -> Unified CM -> EX60(2)-> Share presentation EX60(1)->Unified CM->Blind call transfer -> EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG025	Blind call transfer in EX60 during presentation sharing registered with Cisco VCS	To verify whether EX60(1) which is sharing presentation with EX60(2) can do a blind transfer to EX90 registered with Cisco TelePresence Video Communication Server	EX60(1) -> Cisco VCS -> EX60(2)-> Share presentation EX60(1)-> Cisco VCS-> Blind call transfer -> EX90	Passed	
UCJ10SSVCSG026	Consultative call transfer in EX60 during presentation sharing registered with Cisco VCS	To verify whether EX60(1) which is sharing presentation with EX60(2) can do a Consultative transfer to EX90 registered with Cisco TelePresence Video Communication Server	EX60(1) -> Cisco VCS -> EX60(2)-> Share presentation EX60(1) -> Cisco VCS->Hold / Transfer-> EX90	Passed	
UCJ10SSVCSG027	Audio call between SX20 Quick Set and 7861	To verify whether audio call between Cisco TelePresence SX20 Quick Set and Cisco IP Phone 7861 registered with Cisco Unified Communications Manager works successfully	SX20 Quick Set ->Unified CM->7861	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG028	Inter Cluster audio call between SX20 Quick Set and 7861	To verify if inter cluster audio call between Cisco TelePresence SX20 Quick Set and Cisco IP Phone 7861 registered with Cisco Unified Communications Manager works successfully	SX20 Quick Set ->Unified CM(Cluster 1)->SIP Trunk->Unified CM(Cluster 2)->7861	Passed	
UCJ10SSVCSG029	Hold and resume the audio call between SX20 Quick Set and 7861	To verify if hold and resume the audio call between Cisco TelePresence SX20 Quick Set and Cisco IP Phone 7861 registered with Cisco Unified Communications Manager works successfully	SX20 Quick Set ->Unified CM(Cluster 1)->SIP Trunk->Unified CM(Cluster 2)->7861 SX20 Quickset -> Hold / Resume	Passed	
UCJ10SSVCSG030	Audio Conference between SX20 Quick Set , 7861 and 7925	To Verify if audio conference between SX20 Quickset, 7861 and 7925 registered with Cisco Unified Communications Manager works fine	SX20 Quick Set -> Unified CM->7925-> 7861 -> Audio Conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SVCSG031	Audio call between 7861 and SX20 Quick Set where SX20 Quick Set is registered with Cisco VCS	To verify if audio call between Cisco TelePresence SX20 Quick Set registered with Cisco TelePresence Video Communication Server and Cisco IP Phone 7861 registered with Cisco Unified Communications Manager works successfully	SX20 Quick Set ->Cisco VCS->SIP Trunk->Unified CM->7861	Passed	
UCJ105SVCSG032	Hold and Resume the audio call between 7861 and SX20 Quick Set where SX20 Quick Set is registered with Cisco VCS	To verify if hold and resume the audio call between Cisco TelePresence SX20 Quick Set registered with Cisco TelePresence Video Communication Server and Cisco IP Phone 7861 registered with Cisco Unified Communications Manager works successfully	SX20 Quick Set -> Cisco VCS->SIP Trunk-> Unified CM-> 7861->Hold / Resume	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG033	Audio Conference between SX20 Quick Set , 7861 and 7925 where 7925 login with EM User	To verify if audio Conference between Cisco TelePresence SX20 Quick Set registered with Cisco VCS and Cisco IP Phone 7861, Cisco IP Wireless phone 7925(login with EM user) registered with Cisco Unified Communications Manager works successfully	SX20 Quick Set ->Cisco VCS->SIP Trunk->Unified CM->7925(EM User)->7861->Audio Conference	Passed	
UCJ10SSVCSG034	Audio call between 500-32 and 7841 via ICT Trunk	To verify whether audio call between Cisco TelePresence System 500-32 registered in Cisco Unified Communications Manager Cluster-1 and Cisco Unified IP Phone 7841 registered in Cisco Unified Communications Manager Cluster-2 via ICT trunk works successfully	500-32->Unified CM Cluster1->ICT Trunk->Unified CM Cluster2 ->7841	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105S.VCSG.035	Audio conference call between 500-32 and 69XX IP phones	To verify whether conference call between Cisco TelePresence System 500-32 , Cisco Unified IP Phones 6921 and 6961 registered in Cisco Unified Communications Manager works successfully	500-32->Unified CM -> 6961 -> add participant in 500-32 -> 6921	Passed	
UCJ105S.VCSG.036	Consultative call transfer to 500-32 via SIP trunk	To verify the consultative call transfer from Cisco Unified IP Phone 7841, registered in Cisco Unified Communications Manager cluster1 to Cisco TelePresence System 500-32 registered in Cisco Unified Communications Manager cluster2 works successfully	7841 -> Unified CM cluster1 -> 8941-> Hold/Transfer -> Unified CM cluster1 -> SIP Trunk -> Unified CM cluster2 -> 500-32	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG037	Conference call consultative transfer to 500-32	To verify whether conference call consultative transfer from Cisco Unified IP Phone 8941(1) which is in conference with Cisco Unified IP Phone 8941(2) and 6941 to Cisco TelePresence System 500-32 registered in Cisco Unified Communications Manager works fine	8941(1) -> Unified CM -> 8941(2) ->6941 -> conference -> Hold/Transfer ->500-32	Passed	
UCJ10SSVCSG038	Remote annotation in TX9000	To verify whether Cisco TelePresence TX9000 can annotate shared presentation from Cisco TelePresence System 500-32 registered with Cisco Unified Communications Manager	500-32 -> Unified CM -> TX9000	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG039	Remote annotation in TX9000 after disconnecting presentation source from 500-32	To verify whether Cisco TelePresence TX9000 can annotate shared annotation from Cisco TelePresence System 500-32 after disconnecting presentation source from Cisco TelePresence System 500-32 registered with Cisco Unified Communications Manager	500-32 -> Unified CM -> TX9000	Passed	
UCJ10SSVCSG040	Remote annotation from a transferred call after disconnecting presentation source	Cisco TelePresence System 500-32 shares annotation with DX650 and the call is transferred from DX650 to Cisco TelePresence TX9000 after disconnecting presentation source from Cisco TelePresence System 500-32 registered with Cisco Unified Communications Manager and TX9000 annotate the remote presentation.	500-32 -> Unified CM -> DX650 -> Transfer -> Unified CM -> TX9000	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG041	Remote annotation from a transferred call via SIP trunk	Cisco TelePresence System 500-32(1) registered with Cisco Unified Communications Manager cluster1 shares presentation with DX650 registered with Cisco Unified Communications Manager cluster1 and call is transferred from DX650 to Cisco TelePresence system 500-32(2) registered with Cisco Unified Communications Manager cluster2 which annotate the remote presentation.	500-32(1) -> Unified CM cluster1 -> DX650 -> Transfer -> Unified CM cluster1 -> SIP trunk -> Unified CM cluster2 -> 500-32(2)	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105S.VCSG.042	Remote annotation in TX9000 from 500-32 during conference with DX650 Using MCU 4510	Cisco TelePresence System 500-32 shares presentation with DX650 which adds Cisco TelePresence TX9000 registered with Cisco Unified Communications Manager through conference using MCU 4510 and TX9000 annotate the remote presentation	500-32 -> DX650 -> TX9000->Unified CM->SIP Trunk->MCU 4510->Conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG043	Remote annotation in TX9000 from 500-32 after disconnecting presentation source during conference with DX650 via SIP trunk	Cisco TelePresence System 500-32 registered with Cisco Unified Communications Manager cluster-1 annotate presentation and shares annotation after disconnecting presentation source with DX650 registered with Cisco Unified Communications Manager Cluster-1 which adds Cisco TelePresence TX9000 registered with Cisco Unified Communications Manager Cluster-2 through conference and TX9000 annotate the remote presentation.	500-32 -> Unified CM cluster1 -> DX650 -> Unified CM cluster1 -> SIP trunk -> Unified CM cluster2 -> TX9000->Conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SVCSG044	Sharing annotation after disconnecting presentation source	To verify whether annotation is shared from Cisco TelePresence System 500-32 to Cisco Telepresence TX9000 which are registered with Cisco Unified Communications Manager after disconnecting presentation source from Cisco TelePresence System 500-32	500-32 -> Unified CM -> TX9000	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG045	Share annotation after disconnecting presentation source through SIP trunk	To verify whether annotation is shared from Cisco TelePresence System 500-32(1) registered with Cisco Unified Communications Manager cluster1 to Cisco TelePresence System 500-32(2) registered with Cisco Unified Communications Manager cluster2 after disconnecting presentation source from Cisco TelePresence System 500-32(1)	500-32(1) -> Unified CM Cluster1 -> SIP trunk -> Unified CM cluster2 -> 500-32(2)	Passed	
UCJ10SSVCSG046	Inter Cluster audio call between endpoints registered with Cisco VCS via Cisco VCS Expressway	To verify if the user can Make an Inter Cluster audio call between endpoints registered with Cisco TelePresence Video Communication Server via Cisco TelePresence Video Communication Server Expressway	SX20 Quick Set -> Cisco VCS (Site A)-> Cisco VCS Expressway -> Cisco VCS (Site B) -> EX90-> Audio	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG047	Presentation Sharing between endpoints registered with Cisco VCS via Cisco VCS Expressway	To verify if the user can share the presentation during inter cluster video call between the end points registered with Cisco TelePresence Video Communication Server via Cisco TelePresence Video Communication Server Expressway	SX20 Quick Set-> Cisco VCS (Site A) -> Cisco VCS Expressway -> Cisco VCS (Site B) ->EX60 -> presentation sharing	Passed	
UCJ10SSVCSG048	Presentation Sharing between SX20 Quick Set and Integrator Package C90 registered with Cisco VCS via Cisco VCS Expressway	To verify if the user can share the presentation during inter cluster video call between the end points registered with Cisco TelePresence Video Communication Server via Cisco TelePresence Video Communication Server Expressway	SX20 Quick Set->Cisco VCS (Site A) -> Cisco VCS Expressway -> Cisco VCS (Site B) -> Integrator Package C90 -> presentation sharing	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG049	Inter cluster call transfer between video end points registered with Cisco VCS via Cisco VCS Expressway	To verify that the user can make an inter cluster video call from Cisco TelePresence SX20 Quick Set to Cisco TelePresence System EX60 registered with Cisco Video Communication Server and transfer the call from EX60 to Cisco TelePresence System EX90 via Cisco VCS Expressway	SX20 Quick Set -> Cisco VCS (Site B) -> Cisco VCS Expressway-> Cisco VCS (Site A) -> EX60 -> transfer -> EX90	Passed	
UCJ10SSVCSG050	Consultative transfer from Integrator Package C90 to SX20 Quick Set registered with Cisco VCS via Cisco VCS Expressway	To verify the user can make a call from Cisco TelePresence System EX90 to Integrator Package C90 , hold and transfer the call in C90 to Cisco TelePresence SX20 Quick Set registered with Cisco Video Communication Server via Cisco VCS Expressway	EX90 -> Cisco VCS (Site A) -> Integrator Package C90 -> hold/transfer -> Cisco VCS Expressway -> Cisco VCS (Site B) -> SX20 Quick Set	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG051	Call forward all information display in EX60(2) when call forward all is set to EX60(2) for EX60(1)	Check forwarded Call Information for EX60(2) registered with Cisco Unified Communications Manager when call forward all is set to EX60(2) for EX60(1) registered with Cisco Unified Communications Manager during call from EX90 registered with Cisco Unified Communications Manager to EX60(1)	EX60(2) -> Unified CM -> forwards -> EX90	Failed	CSCuo78101
UCJ10SSVCSG053	Check the hold status of participant during conference	Check the hold status of EX60(1) registered with Cisco Unified Communications Manager in the conference call with EX60(2) and EX90 which are registered with Cisco Unified Communications Manager	EX60(1) -> Unified CM -> EX60(2) -> add participant -> EX90	Failed	CSCuo88112

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSVCSG055	Transfer a call during scheduled conference	Verify whether Cisco TelePresence System EX90 registered in Cisco Telepresence Video Communications Server can transfer a call to Cisco TelePresence System EX60 when it is in scheduled conference managed by Cisco TelePresence Management Suite	Cisco TMS-> Cisco VCS-> EX90-> EX60(1)-> SX20 Quick Set-> Conference EX90 -> Transfer -> Cisco VCS->EX60(2)	Failed	CSCuo90595
UCJ10SSVCSG056	Participant List after a directed call park in TX9000	Verify whether participant list of Cisco TelePresence TX9000 shows the correct participants after a directed call park in Cisco Unified IP phone 8945	TX9000-> Unified CM-> EX90-> Add Participant -> Unified CM -> 8945 ; 8945(1)-> Directed Call Park-> Unified CM -> 8945(2) -> Retrieve	Failed	CSCuo99704
UCJ10SSVCSG057	Park number in EX90 after a directed call park	Verify whether park number shows correctly in Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager when Cisco Unified IP phone 8945 does a directed call park	EX90->Unified CM->8945 ; 8945->Directed Call Park	Failed	CSCuo87291

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.VCSG.058	Transfer a inter cluster call from 7925 wireless phone to EX60	Verify whether user can transfer a inter cluster call from 7925 wireless phone to EX60 registered with Unified CM	7925-> Unified CM Cluster 1-> SIP Trunk-> Unified CM Cluster 2 -> EX90 7925 -> Unified CM Cluster-1 -> Transfer -> Unified CM Cluster -2 -> EX60	Failed	CSCuo47595

Cisco Jabber for iPhone and iPad

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CJIG.001	Make an inter-cluster call from Jabber for iPhone to Jabber for iPad via SIP Trunk	Verify whether the Jabber for iPhone can make video call to Jabber for iPad successfully and after placing the call from Jabber for iPad Presence status should change into "on a call" in Jabber for iPhone contact list	CJI -> Unified CM1 -> SIP Trunk -> Unified CM2 -> CJiPad	Passed	
UCJ10.5S.CJIG.003	Make an inter-cluster call from Jabber for iPhone to DX650 IP Phone via Cisco Unified Border Element through SIP Trunk	Verify whether the Jabber for iPhone can make video call to DX650 IP Phone and DX650 IP Phone should attend the video call successfully	CJI -> Unified CM1-> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> DX650	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CJI.G.004	Make an inter-cluster voice call from Jabber for iPhone to 8831 IP Phone via ICT Trunk	Verify whether the Jabber for iPhone can make voice call to 8831 IP Phone and 8831 IP Phone should attend the voice call successfully	CJI -> Unified CM1 -> ICT Trunk -> Unified CM2 -> 8831	Passed	
UCJ10.5S.CJI.G.007	Check Jabber for iPhone Version is displayed in Unified CM	Verify whether the Jabber for iPhone version is displayed in Unified CM successfully	NA	Passed	
UCJ10.5S.CJIPad.G.001	Make an inter-cluster call from Jabber for iPad to Jabber for Android via ICT Trunk	Verify whether the Jabber for iPad can make video call to Jabber for Android and Jabber for Android should attend the video call successfully	CJiPad -> Unified CM1 -> ICT Trunk -> Unified CM2 -> CJA	Passed	
UCJ10.5S.CJIPad.G.003	Check Registered Status of Jabber for iPad is displayed in Unified CM	Verify whether the Registered Status of Jabber for iPad is displayed successfully in Unified CM	NA	Passed	
UCJ10.5S.CJIPad.G.005	Make an inter-cluster voice call from Jabber for iPad to 8831 IP Phone via ICT Trunk	Verify whether the Jabber for iPad can make voice call to 8831 IP Phone and 8831 IP Phone attends the voice call successfully	CJiPad -> Unified CM1 -> ICT Trunk -> Unified CM2 -> 8831	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CJIPadG.007	Make an inter-cluster voice call from Jabber for iPad to 8831 IP Phone via Cisco Unified Border Element through SIP Trunk	Verify whether the Jabber for iPad can make voice call to 8831 IP Phone and 8831 IP Phone attends the voice call successfully	CJiPad -> Unified CM1 -> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> 8831	Passed	
UCJ10.5S.CJI.G.007	Check Jabber for iPhone Version is displayed in Unified CM	Verify that the Jabber for iPhone Version is displayed successfully in Unified CM	NA	Passed	
UCJ10.5S.CJI.G.009	Check Jabber for iPhone can make a call in chat option	Verify that the Jabber for iPhone can make a call in chat option successfully	NA	Passed	
UCJ10.5S.CJI.G.011	Check Message Notification is displayed on the iPhone when Jabber was running in the background	Verify that the Message Notification is displayed on the iPhone when Jabber was running in the background	NA	Passed	
UCJ10.5S.CJI.G.014	Check Call Notification is displayed on the iPhone when Jabber was running in the background	Verify that the Call Notification is displayed on the iPhone when Jabber was running in the background	NA	Passed	
UCJ10.5S.CJIPadG.009	Check Message Notification is displayed on the iPad when Jabber was running in the background	Verify that the Message Notification is displayed on the iPad when Jabber was running in the background	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5SCJIPadG.011	Check Call Notification is displayed on the iPad when Jabber was running in the background	Verify that the Call Notification is displayed on the iPad when Jabber was running in background	NA	Passed	
UCJ10.5SCJIPadG.012	Check Jabber for iPad is changing the Presence status after call was connected to Jabber for iPhone via SIP Trunk	Verify that Jabber for iPad is changing the Presence status after call was connected to Jabber for iPhone via SIP Trunk	CJiPad -> Unified CM1 -> SIP Trunk -> Unified CM2 -> CJI	Passed	
UCJ10.5SCJIPadG.007	Make an inter-cluster voice call from Jabber for iPad to 8831 IP Phone via Cisco Unified Border Element through SIP Trunk	Verify whether the Jabber for iPad can make voice call to 8831 IP Phone and 8831 IP Phone attends the voice call successfully	CJiPad -> Unified CM1 -> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> 8831	Passed	

Cisco Jabber for Android

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5SCJAG.001	Check the Jabber for Android Registered status is displayed in Unified CM	Verify that the Jabber for Android Registered status is displayed successfully in Unified CM	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5SCJAG.003	Check the Jabber for Android Version is displayed correctly in Unified CM	Verify that the Jabber for Android Version is displayed correctly in Unified CM	NA	Passed	
UCJ10.5SCJAG.005	Make an inter-cluster call from Jabber for Android to Jabber for iPad via ICT Trunk	Verify that the Jabber for Android can make video call to Jabber for iPad and after placing the call from Jabber for iPad the Presence status should change into "on a call" in Jabber for Android contact list	CJA -> Unified CM1 ->ICT Trunk -> Unified CM2 -> CJiPad	Passed	
UCJ10.5SCJAG.007	Make an inter-cluster voice call from Jabber for Android to 9971 IP Phone via SIP Trunk	Verify that the Jabber for Android can make voice call to 9971 IP Phone and 9971 IP Phone should attend the voice call successfully	CJA ->Unified CM1 ->SIP Trunk -> Unified CM2 ->9971	Passed	
UCJ10.5SCJAG.009	Check Jabber for Android can make a call in chat option	Verify that the Jabber for Android can make a call in chat option successfully	NA	Passed	
UCJ10.5SCJAG.011	Check Jabber for Android Missed Call Notification should come in Call History after disconnecting the call	Verify that the Jabber for Android Missed Call Notification comes in Call History successfully after disconnecting the call	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5SCJAG.012	Check Automatic-away is working in Jabber for Android	Verify that the Automatic-away is working successfully in Jabber for Android	NA	Passed	
UCJ10.5SCJAG.014	Check Japanese keypad is working in Jabber for Android when chatting with members	Verify that the Japanese keypad is working in Jabber for Android when chatting with members	NA	Passed	

Cisco Jabber for Windows

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5SCJWG.158	Conference between CJA and CJW	Verify that Cisco Jabber for Windows A successfully initiates conference call with Cisco Jabber for Windows B and Cisco Jabber for Android	CJW A -> Unified CM -> CJA -> Unified CM -> CJW B	Passed	
UCJ10.5SCJWG.162	Desktop share after complete transfer	Verify that Cisco Jabber for Windows A successfully shares the desktop to Cisco Jabber for Windows B after complete transfer is made	CJW A -> Unified CM -> Phone B -> Unified CM -> CJW B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCJWG160	Call made via CUBE from CJW to IP phone present in shared line	Verify that Cisco Jabber for Windows makes call via CUBE to IP phone present in shared line successfully	CJW A -> Unified CM1 -> H.323 -> CUBE -> H.323 -> Unified CM2 -> Phone B	Passed	
UCJ10SSCJWG122	Transfer made by CJW installed in Windows 8.1 to 8831 present in shared line via ICT trunk	Verify that Cisco Jabber for Windows B installed in Windows 8.1 platform completely transfers the incoming call from CJW A to 8831 IP phone present in shared line with IP phone D via ICT trunk successfully	CJW A -> Unified CM1 -> CJW B -> Unified CM1 -> ICT Trunk -> Unified CM2 -> 8831	Passed	
UCJ10SSCJWG129	Call made from Cisco Jabber for Windows to DX650 and check call transfer	Verify that Cisco Jabber for Windows installed in Windows 8.1 platform makes call to DX650 and transfers the call completely to IP phone D	CJW A -> Unified CM -> DX650 -> Unified CM -> Phone D	Passed	
UCJ10SSCJWG136	Audio/Visual notification in CJW	Verify that CJW B receives visual/audio notification when it receives incoming call within the cluster	Phone A -> Unified CM -> CJW B -> Unified CM -> CJW A	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCJWG.147	Visual notification in CJW when receiving call while doing multiple transfer within cluster	Verify that Cisco Unified IP phone D makes call to IP phone C and IP phone C attends and transfers the call to CJW and check whether the Cisco Jabber for Windows receives visual notification when it receives incoming call from IP phone C successfully	Phone D -> Unified CM -> Phone C -> Unified CM -> CJW A -> Unified CM -> Phone B	Passed	
UCJ10SSCJWG.132	Call made from CJW A to CJW B via CUBE and check directed call park in CJW	Verify that Cisco Jabber for Windows A makes call to Cisco Jabber for Windows B via CUBE through H.323 to H.323 interworking and check whether the directed call park in CJW B is made successfully	CJW A -> Unified CM1 -> H.323 -> CUBE -> H.323 -> Unified CM2 -> CJW B -> Unified CM2 -> Phone B	Passed	
UCJ10SSCJWG.152	Desktop share between Cisco TelePresence SX20 Quick Set and Cisco Jabber for Windows after establishing the call	Verify that Cisco Jabber for Windows successfully shares its desktop with Cisco TelePresence SX20 Quick Set after the call is established between Cisco TelePresence SX20 Quick Set and CJW	SX20 -> Unified CM -> CJW A	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SCJWG157	Group chat between two Cisco Jabber for Windows and Cisco Jabber for Android	Verify that group chat is successfully initiated between Cisco Jabber for Windows A and Cisco Jabber for Android by CJW B	NA	Passed	
UCJ105SCJWG159	Check for the Group Chat during Conference by clicking the group chat icon	Verify that the Cisco Jabber for Windows A can successfully initiate conference call with Cisco Jabber for Windows B and Cisco Jabber for Android and click the "group chat" icon and check whether the group chat is initiated automatically	CJW A -> Unified CM -> CJA -> Unified CM -> CJW B	Passed	
UCJ105SCJWG121	Consult transfer by CJW in Windows 8.1 platform via SIP trunk	Verify that the Cisco Jabber for Windows A can make call to Cisco Jabber for Windows B and CJW B transfers the call to IP phone B and does consult transfer via SIP trunk successfully	CJW A -> Unified CM1 -> CJW B -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCJWG123	Video call establishment between Jabber clients using Windows 8.1 platform via CUBE	Verify that the video call is established successfully between two Cisco Jabber for Windows using Windows 8.1 platform via CUBE through H.323 to H.323 interworking	CJW A -> Unified CM1 -> H.323 -> CUBE -> H.323 -> Unified CM2 -> CJW B	Passed	
UCJ10SSCJWG130	Multiple transfer made by CJW using Windows 8.1 platform	Verify that Cisco Jabber for Windows does multiple transfer successfully	Phone B -> Unified CM -> CJW A -> Unified CM -> Phone C -> Unified CM -> Phone D	Passed	

Cisco Jabber for Mac

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCJMG001	Video Call from Jabber for Mac to Jabber for iPhone via SIP Trunk	Verify that Cisco Jabber for Mac Client can able to make Video Call to Cisco Jabber for iPhone via SIP trunk in different cluster	Jabber for Mac -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Jabber for iPhone	Passed	
UCJ10SSCJMG002	Video Call from Jabber for Mac to Jabber for iPhone via ICT Trunk	Verify that Cisco Jabber for Mac Client can able to make Video Call to Cisco Jabber for iPhone via ICT trunk in different cluster	Jabber for Mac -> Unified CM1 -> ICT Trunk -> Unified CM2 -> Jabber for iPhone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SCJMG003	Video Call from Jabber for Mac to Jabber for iPhone in different cluster via Cisco Unified Border Element through SIP Trunk	Verify that Cisco Jabber for Mac Client can able to make Video Call to Cisco Jabber for iPhone via Cisco Unified Border Element through SIP Trunk	Jabber for Mac -> Unified CM1 -> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> Jabber for iPhone	Passed	
UCJ105SCJMG004	Video Call from Jabber for Mac to Jabber for iPhone in different cluster via Cisco Unified Border Element through ICT Trunk	Verify that Cisco Jabber for Mac Client can able to make Video Call to Cisco Jabber for iPhone via Cisco Unified Border Element through ICT Trunk	Jabber for Mac -> Unified CM1 -> ICT Trunk -> CUBE- > ICT Trunk-> Unified CM2 -> Jabber for iPhone	Passed	
UCJ105SCJMG005	Video Call from Jabber for Mac to Jabber for iPhone in different cluster via Cisco Unified Border Element through SIP-ICT Trunk	Verify that Cisco Jabber for Mac Client can able to make Video Call to Cisco Jabber for iPhone via Cisco Unified Border Element through SIP-ICT Trunk	Jabber for Mac -> Unified CM1 -> SIP Trunk -> CUBE -> ICT Trunk -> Unified CM 2 -> Jabber for iPhone	Passed	
UCJ105SCJMG006	Video Call from Jabber for Mac to Jabber for iPhone in different cluster via Cisco Unified Border Element through ICT-SIP Trunk	Verify that Cisco Jabber for Mac Client can able to make Video Call to Cisco Jabber for iPhone in different cluster via Cisco Unified Border Element through ICT-SIP Trunk	Jabber for Mac -> Unified CM1 -> ICT Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> Jabber for iPhone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SCJMG007	Video Call from Jabber for Mac to Jabber for iPhone in same cluster	Verify that Cisco Jabber for Mac Client can able to make Video Call to Cisco Jabber for iPhone in same cluster	Jabber for Mac -> Unified CM1 -> Jabber for iPhone	Passed	
UCJ105SCJMG008	Video Call from Jabber for Mac to Jabber for iPhone via PSTN	Verify that Cisco Jabber for Mac Client can able to make Video Call to Cisco Jabber for iPhone via PSTN	Jabber for Mac -> Unified CM1 -> PSTN -> Unified CM2 -> Jabber for iPhone	Passed	
UCJ105SCJMG009	Chat history until logout when Jabber for Mac do Instant Messaging with Jabber for iPhone in same cluster	Verify when Jabber for Mac do Instant Messaging with Jabber for iPhone in same cluster and after logout the chat messages of Jabber for Mac should not be there in the history	Jabber for Mac -> Unified CM1 -> Jabber for iPhone	Passed	
UCJ105SCJMG010	Chat history until logout when Jabber for Mac do Instant Messaging with Jabber for iPhone in different cluster	Verify when Jabber for Mac do Instant Messaging with Jabber for iPhone in different cluster and after logout the chat messages of Jabber for Mac should not be there in the history	Jabber for Mac -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Jabber for iPhone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SCJMG011	Chat history when Jabber for Mac close the chat box after Instant Messaging with Jabber for iPhone in same cluster	Verify when Jabber for Mac do Instant messaging with Jabber for iPhone in same cluster and close the chat box, the chat messages should be present in the history	Jabber for Mac -> Unified CM1 -> Jabber for iPhone	Passed	
UCJ105SCJMG012	Chat history when Jabber for Mac close the chat box after Instant Messaging with Jabber for iPhone in different cluster	Verify when Jabber for Mac do Instant messaging with Jabber for iPhone in different cluster and close the chat box, the chat messages should be present in the history	Jabber for Mac -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Jabber for iPhone	Passed	
UCJ105SCJMG013	Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via SIP Trunk	Verify that Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via SIP Trunk	Jabber for Mac -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Jabber for iPhone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCIMG014	Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via ICT Trunk	Verify that Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via ICT Trunk	Jabber for Mac -> Unified CM1 -> ICT Trunk -> Unified CM2 -> Jabber for iPhone	Passed	
UCJ10SSCIMG015	Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via Cisco Unified Border Element through SIP Trunk	Verify that Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via Cisco Unified Border Element through SIP Trunk	Jabber for Mac -> Unified CM1 -> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> Jabber for iPhone	Passed	
UCJ10SSCIMG016	Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via Cisco Unified Border Element through ICT Trunk	Verify that Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via Cisco Unified Border Element through ICT Trunk	Jabber for Mac -> Unified CM1 -> ICT Trunk -> CUBE -> ICT Trunk -> Unified CM2 -> Jabber for iPhone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SCJMG017	Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via Cisco Unified Border Element through SIP-ICT Trunk	Verify that Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via Cisco Unified Border Element through SIP-ICT Trunk	Jabber for Mac -> Unified CM1 -> SIP Trunk -> CUBE -> ICT Trunk -> Unified CM2 -> Jabber for iPhone	Passed	
UCJ105SCJMG018	Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via Cisco Unified Border Element through ICT-SIP Trunk	Verify that Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via Cisco Unified Border Element through ICT-SIP Trunk	Jabber for Mac -> Unified CM1 -> ICT Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> Jabber for iPhone	Passed	
UCJ105SCJMG019	Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone in same cluster	Verify that Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone in same cluster	Jabber for Mac -> Unified CM1 -> Jabber for iPhone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5SCIMG020	Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via PSTN	Verify that Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call is connected with Jabber for iPhone via PSTN	Jabber for Mac -> Unified CM1 -> PSTN -> Unified CM2 -> Jabber for iPhone	Passed	
UCJ10.5SCIMG021	Jabber for Mac does Group chat with other Jabber clients in same cluster	Verify that Jabber for Mac does Group chat with other Jabber clients in same cluster and Group chat invitation does not contain chat topic	Jabber for Mac -> Unified CM1 -> Jabber Client	Passed	
UCJ10.5SCIMG022	Jabber for Mac does Group chat with other Jabber clients in different cluster	Verify that Jabber for Mac does Group chat with other Jabber clients in different cluster and Group chat invitation does not contain chat topic	Jabber for Mac -> Unified CM1 -> Unified CM2 -> Jabber Client	Passed	
UCJ10.5SCIMG023	Jabber for Mac does Group Chat Conversation with other Jabber clients in same cluster	Verify that Jabber for Mac does Group chat with other Jabber clients in same cluster and Group Chat Conversation does not appear in the Chat history	Jabber for Mac -> Unified CM1 -> Jabber Client	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCJMG024	Jabber for Mac does Group Chat Conversation with other Jabber clients in different cluster	Verify thatmJabber for Mac does Group chat with other Jabber clients in different cluster and Group Chat Conversation does not appear in the Chat history	Jabber for Mac -> Unified CM1 -> Unified CM2 -> Jabber Client	Passed	
UCJ10SSCJMG025	In Jabber for Mac chat history, input one name to filter out the chat history with the user in same cluster	Verify that in Jabber for Mac Chat history, user can able to input one name to filter out the chat history with the user in same cluster	Jabber for Mac -> Unified CM1 -> Jabber Client	Passed	
UCJ10SSCJMG026	In Jabber for Mac chat history, input one name to filter out the chat history with the user in different cluster	Verify that in Jabber for Mac Chat history, user can able to input one name to filter out the chat history with the user in different cluster	Jabber for Mac -> Unified CM1 -> Unified CM2 -> Jabber Client	Passed	
UCJ10SSCJMG027	Always start calls with Video in Jabber for Mac when Jabber for iPhone calls via SIP Trunk	Verify that Jabber for Mac starts calls with Video when Jabber for iPhone in different cluster make call to Jabber for Mac via SIP trunk	Jabber for iPhone -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Jabber for Mac	Passed	

Cisco Unified Border Element

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCUBEG201	Make call from one site to another site when configuring "voice class dpg" which refer the dial-peer with preference 1 on CUBE	Verify whether the call from one site to another site when configuring "voice class dpg" which refer the dial-peer with preference 1 is successful	Phone A -> Unified CM1 -> CUBE -> Unified CM2 -> Phone B	Passed	
UCJ10SSCUBEG202	Call between one site to another site through redundant dial-peer when primary is not present	Verify that the call between one site to another site through redundant dial-peer when primary is not present is successful	Phone A -> Unified CM1 -> CUBE -> Unified CM2 -> Phone B	Passed	
UCJ10SSCUBEG203	Call should be successful when Incoming dial-peer has the mapping of outgoing dial-peer on CUBE	Verifying the Configurability of destination calling e164-pattern-map	Phone A -> Unified CM1 -> CUBE -> Unified CM2 -> Phone B	Passed	
UCJ10SSCUBEG205	Call between one site to another site through H.323 - H.323 Calls with configuration of "voice class provision policy"	Verify the call between one site to another site through H.323 with configuration of "voice class provision policy"	Phone A -> Unified CM1 -> H.323 -> CUBE -> H.323 -> Unified CM2 -> Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCUBEG300	Basic video call between Site 1 Phone to another Site2 Phone through SIP trunk using CUBE with codec Renegotiation	Basic video call between Site 1 Phone to another Site2 Phone through SIP trunk using CUBE with codec Renegotiation	Phone A -> Unified CM1 -> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> Phone B	Passed	
UCJ10SSCUBEG302	Basic video call between Phone in one site to Phone in other site with codec Renegotiation	Basic video call between Phone in one site to phone in another site with codec Renegotiation - Both Video(H264) & Audio(MP4A-LATM)	Phone A -> Unified CM1 -> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> Phone B	Passed	
UCJ10SSCUBEG305	Basic video call between Phone in one site to Phone in other site with codec Renegotiation-EO Re-INVITE is received after the EO INVITE is sent	DO-EO video call with codec Renegotiation-EO Re-INVITE is received after the EO INVITE is sent	Phone A -> Unified CM1 -> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> Phone B	Passed	
UCJ10SSCUBEG307	Holding and Retrieving the other site calls	Verify that the call hold functionality is working successfully for incoming calls from other sites after configuring the SIP Early offer Renegotiation on CUBE	Phone A -> Unified CM1 -> Phone B -> Unified CM1 -> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> Phone D	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUBEG306	Basic video call transfer between one site to another site after configuring the SIP Early offer Renegotiation on CUBE	Verify that the basic video call transfer between one site to another site after configuring the SIP Early offer Renegotiation on CUBE is successful	Phone A -> Unified CM1 -> Phone B -> Unified CM1 -> SIP Trunk -> CUBE -> SIP Trunk -> Unified CM2 -> Phone D	Passed	

Cisco Unified Survivable Remote Site Telephony

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSRSTG104	Multiple transfer using 7821 IP phone	Verify that the multiple transfer works successfully by using 7821 IP phone	Phone A -> Unified SRST -> 7821 IP Phone -> Phone B -> Phone D	Passed	
UCJ10SSRSTG103	Add 7821 IP phone in hunt group and check by establishing call with 7821 IP phone from other IP phone	Verify that the Cisco Unified IP phone 7821 is added in the hunt group and call has made from IP phone D to hunt pilot number and also check whether 7821 IP phone attends the call successfully	Phone D -> Unified SRST -> Hunt pilot number -> 7821 IP Phone	Passed	
UCJ10SSRSTG105	Call made via PSTN from Unified CM2 to IP phone B in Unified SRST and call transferred to IP phone C	Verify that the Cisco Unified IP phone A in Unified CM makes call via PSTN to IP phone B in SRST mode and Phone B transfers the call to IP phone C successfully	Phone A -> Unified CM2 -> PSTN -> Unified SRST -> Phone B -> Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SSRSTG106	Forwarding incoming call via PSTN to IP phone in Unified SRST	Verify that the call forward is set via PSTN in IP phone B and check call gets forwarded to IP phone C present in Unified CM2 via PSTN successfully	Phone A -> Unified SRST -> Phone B -> Unified SRST -> PSTN -> Unified CM2 -> Phone C	Passed	
UCJ105SSRSTG107	Initiate call from 7861 IP phone in Unified SRST to other Unified CM	Verify that the 7861 IP phone successfully makes call to IP phone B registered in other Unified CM	7861 IP Phone -> Unified SRST -> PSTN -> Unified CM -> Phone B	Passed	
UCJ105SSRSTG108	Call made to hunt pilot number in Unified SRST via PSTN	Verify that Cisco Unified IP phone D from Unified CM2 makes call to hunt pilot number in Unified SRST successfully and IP phone B present in hunt group attends the call	Phone D -> Unified CM2 -> PSTN -> Unified SRST -> Hunt pilot number	Passed	
UCJ105SSRSTG109	Call establishment between two 78xx IP phones in Unified SRST	Verify that Cisco Unified IP phone 7821 makes call to 7861 IP phone and check call gets established successfully in Unified SRST	7821 IP Phone -> Unified SRST -> 7861 IP Phone	Passed	

Cisco Unified Communications IM and Presence

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCCMIMPR009	Check for the compatibility of IM and Presence Admin Page in IE 10	Verify whether the IM and Presence Admin Page is working successfully in IE10	NA	Passed	
UCJ10SCCMIMPR011	Check for the compatibility of Certificate Management Page after upgrading to IE10	Verify whether the certificate Management Page is working successfully after upgrade to IE10	NA	Passed	
UCJ10SCCMIMPR023	Uploading Oracle database certificate in cup-xmpp-trust	Verify Oracle database certificate is uploaded successfully in cup-xmpp-trust	NA	Passed	
UCJ10SCCMIMPR025	Check Oracle database certificate is assigned to Message Archiver Service	Verify Oracle database certificate is successfully assigned to Message Archiver Service	NA	Passed	
UCJ10SCCMIMPR008	Check sorting and filtering is working fine when tapping the Find button	Verify that the sorting and filtering is working successful when tapping the Find button	NA	Passed	
UCJ10SCCMIMPR012	Check for the display of version settings in IE10 after changing into JP Language	Verify that the version settings are displayed successfully in IE10 after changing into JP language	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCCMIMRG021	Check for the display of correct Japanese message when starting the services in IM&P Serviceability Page (Japanese) by using IE10	Verify that the correct Japanese message is displayed when starting the services in IM&P Serviceability Page(Japanese) by using IE10	NA	Passed	
UCJ10SCCMIMRG001	Check Multi-Server Option is displayed in the distribution field when clicking the Generate CSR option in the IM&P OS Admin Page	Verify that the Multi-Server Option is displayed in the distribution field when generating CSR	NA	Passed	

Cisco Unity Connection

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCG001	Record a voice message in Web Inbox using SIP URI and play it from Web Inbox	Verify whether the user can able to record a voice message in Web Inbox by specifying SIP URI in phone field and playing it from Web Inbox	Web Inbox -> Unity Connection -> Web Inbox	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCUCG003	Record a voice message in Web Inbox using SIP URI and play it from View Mail for Outlook	Verify whether the user can able to record a voice message in Web Inbox by specifying SIP URI in phone field and playing the voice message in View Mail for Outlook	Web Inbox -> Unity Connection -> Exchange Server -> Outlook 2010 -> View Mail for Outlook	Passed	
UCJ10SSCUCG004	Record a voice message in Web Inbox using SIP URI and play it from Visual Voicemail	Verify whether the user can able to record a voice message in Web Inbox by specifying SIP URI in phone field and listening it from Unified IP Phone Visual Voicemail application	Web Inbox -> Unity Connection -> Unified CM -> Unified IP Phone -> Visual Voicemail Midlet	Passed	
UCJ10SSCUCG007	Access Web Inbox from web browser of Unified IP Phone DX650 and record a voice message using SIP URI	Verify whether the user can able to access a Web Inbox from DX650 browser and record a voice message by specifying SIP URI in phone field.	Unified IP Phone DX650 -> Web Inbox -> Unity Connection -> Web Inbox	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SCUCG008	Enable SMTP Message Notification for Notification Devices and record a message from Web Inbox using SIP URI	Verify whether the user can able to get Message Notification mail when they receive a voice message from Web Inbox when voicemail sent by using SIP URI in phone field	Web Inbox -> Unity Connection -> Exchange Server -> Outlook 2010	Passed	
UCJ105SCUCG020	Check the mailbox count of user when voicemail is sent using SIP URI from Web Inbox	Verify the user mailbox count in connection server under user settings after they received the voicemail when voicemail is sent using SIP URI from Web Inbox	Web Inbox -> Unity Connection	Passed	
UCJ105SCUCG030	Enable Live Reply to user when user retrieve the voicemail through TUI	Verify the live reply to user is successful after retrieving the voicemail through TUI	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unity Connection -> Unified IP Phone B -> Unity Connection -> Unified CM -> Unified IP Phone A	Passed	
UCJ105SCUCG036	Check REFER BY SIP Message in network trace after enabling Legal Intercept for Live Reply	To verify the SIP Messages while live reply to user after retrieving the voicemail through TUI	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unity Connection -> Unified IP Phone B -> Unity Connection -> Unified CM -> Unified IP Phone A	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCUCG046	Create Distribution list by adding local users and assign them as call handler owner	Verify whether the user can able to create Distribution list by adding local users and can assign them as call handler owner to record greetings	Unified IP Phone -> Unified CM -> Unity Connection	Passed	
UCJ10SSCUCG056	Leave a voicemail to user after configuring Unified Messaging Account with NTLMv2 Authentication	Verify whether the user can able to leave a voicemail after configuring Unified Messaging Account with NTLMv2 Authentication	Unified IP Phone -> Unified CM -> Unity Connection -> Exchange Server -> Outlook 2010	Passed	
UCJ10SSCUCG012	Record a private and secure voice message using SIP URI in Web Inbox	Verify whether the user is able to record a voice message with private and secure message in Web Inbox by specifying SIP URI in phone field	Web Inbox -> Unity Connection -> Web Inbox	Passed	
UCJ10SSCUCG015	Play a voice message in Web Inbox using SIP URI when a voice mail is sent from Visual Voicemail	Verify whether the user can able to play a voice message using SIP URI in Web Inbox when user received the voice mail from Visual Voicemail Midlet application	Visual Voicemail Midlet -> Unity Connection -> Web Inbox	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SCUCG021	Reply to voicemail through TUI when voicemail is sent using SIP URI from Web Inbox	Verify whether the user can able to reply to voicemail through TUI when voicemail is sent using SIP URI from Web Inbox	Web Inbox -> Unity Connection -> Unified CM -> Unified IP Phone -> Unified CM -> Unity Connection	Passed	
UCJ105SCUCG023	Check the Message Status while replying through TUI when urgent and secure voicemail is sent using SIP URI from Web Inbox	Verify the voice message status such as urgent, private and secure is successful while replying to message through TUI	Web Inbox -> Unity Connection -> Unified CM -> Unified IP Phone -> Unified CM -> Unity Connection	Passed	
UCJ105SCUCG029	Check Message Indicators in Web Inbox and Visual Voicemail when voicemail is sent using SIP URI from Web Inbox	Verify for the message indicators in Web Inbox and Visual Voicemail Midlet when user received the urgent, secure and private voice message	Web Inbox -> Unity Connection -> Web Inbox and Visual Voicemail Midlet	Passed	
UCJ105SCUCG031	Enable Live Reply with Supervised Transfer	Verify whether the user can able to Live Reply by Supervised Transfer	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unity Connection -> Unified IP Phone B -> Unity Connection -> Unified CM -> Unified IP Phone A	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCG043	Enable Live Reply to Jabber Client user and verify the behavior in REFER BY SIP Message	Verify whether the user can able to Live Reply to Jabber for Windows user after retrieving the voicemail through TUI and verify the SIP Messages	CJW -> Unified CM -> Unified IP Phone -> Unity Connection -> Unified IP Phone -> Unity Connection -> Unified CM -> CJW	Passed	
UCJ10SCUCG049	Add new users to Distribution List and check for the updation of Call Handler owners	Verify whether the user can able to add new users to Distribution List and checking for the updation of Call Handler owners and record a greeting	Unified IP Phone -> Unified CM -> Unity Connection	Passed	
UCJ10SCUCG057	Listen to calendar appointments through TUI after configuring Unified Messaging Account with NTLMv2 Authentication	Verify whether the user can able to listen to calendar appointments through TUI after configuring Unified Message Account with NTLMv2 Authentication	Unified IP Phone -> Unified CM -> Unity Connection -> Exchange Server	Passed	
UCJ10SCUCG060	Retrieve E-mail through TUI after configuring Unified Messaging Account with NTLMv2 Authentication	Verify whether the user can able to retrieve E-mail through TUI after configuring Unified Messaging Account with NTLMv2 Authentication	Unified IP Phone -> Unified CM -> Unity Connection -> Exchange Server	Passed	

Cisco Unified Attendant Console

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUACG003	Queue color displaying for forced delivery call (Intra)	Verify whether color for force delivery call is displayed correctly in CUAC client GUI page when a call comes to CUAC operator from an Unified IP Phone registered in Unified CM	Unified IP Phone A -> Unified CM -> CUAC Operator -> CUAC (Client)	Passed	
UCJ10SCUACG004	Queue color displaying for forced delivery call (SIP)	Verify whether color for force delivery call is displayed correctly in CUAC client GUI page when a call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via SIP trunk (Transfer)	Unified IP Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Unified IP Phone B -> Unified CM2 -> CUAC Operator -> CUAC (Client)	Passed	
UCJ10SCUACG005	Queue color displaying for forced delivery call (ICT)	Verify whether color for force delivery call is displayed correctly in CUAC client GUI page when a call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via ICT trunk (Transfer)	Unified IP Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Unified IP Phone B -> Unified CM2 -> CUAC Operator -> CUAC (Client)	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCUACG006	Queue color displaying for forced delivery call (PSTN)	Verify whether color for force delivery call is displayed correctly in CUAC client GUI page when a call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via PSTN trunk (Transfer)	Unified IP Phone A -> Unified CM1 -> PSTN -> Unified CM2 -> Unified IP Phone B -> Unified CM2 -> CUAC Operator -> CUAC (Client)	Passed	
UCJ10SSCUACG007	Queue color displaying for forced delivery call (Intra-Call Park)	Verify whether color for force delivery call is displayed correctly after retrieving the parked call in CUAC client GUI page when a call comes to CUAC operator from an Unified IP Phone registered in Unified CM	Unified IP Phone A -> Unified CM -> CUAC Operator -> CUAC (Client)	Passed	
UCJ10SSCUACG008	Queue color displaying for forced delivery call (SIP Park)	Verify whether color for force delivery call is displayed correctly in CUAC client GUI page after retrieving the parked call, when a call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via SIP trunk (Park)	Unified IP Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Unified IP Phone B -> Unified CM2 -> CUAC Operator -> CUAC (Client)	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10_5_SCUAC_G009	Queue color displaying for forced delivery call (ICT Trunk - Park)	Verify whether color for force delivery call is displayed correctly in CUAC client GUI page after retrieving the parked call, when call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via ICT trunk	Phone A -> Unified CM1 -> ICT Trunk -> Unified CM2 -> Phone B -> Unified CM2 -> CUAC operator -> CUAC (Client)	Passed	
UCJ10_5_SCUAC_G010	Queue color displaying for forced delivery call (PSTN - Call Park)	Verify whether color for force delivery call is displayed correctly in CUAC client GUI page after retrieving the parked call, when call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via PSTN trunk	Phone A -> Unified CM1 -> PSTN -> Unified CM2 -> Phone B -> Unified CM2 -> CUAC operator -> CUAC (Client)	Passed	
UCJ10_5_SCUAC_G011	Queue color displaying for Broadcast queue (Intra)	Verify whether color for broadcast queue call is displayed correctly in CUAC client GUI page when call comes to CUAC operator from an Unified IP Phone registered in Unified CM	Phone A -> Unified CM -> CUAC operator -> CUAC (Client)	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCUACG012	Queue color displaying for broadcast call (SIP Trunk)	Verify whether color for broadcast call is displayed correctly in CUAC client GUI page when call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via SIP trunk (Call Transfer)	Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Phone B -> Unified CM2 -> CUAC operator -> CUAC (Client)	Passed	
UCJ10SSCUACG013	Queue color displaying for broadcast call (ICT Trunk)	Verify whether color for broadcast call is displayed correctly in CUAC client GUI page when call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via ICT trunk (Call Transfer)	Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Phone B -> Unified CM2 -> CUAC operator -> CUAC (Client)	Passed	
UCJ10SSCUACG014	Queue color displaying for broadcast call (PSTN)	Verify whether color for broadcast call is displayed correctly in CUAC client GUI page when call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via PSTN trunk (Call Transfer)	Phone A -> Unified CM1 -> PSTN -> Unified CM2 -> Phone B -> Unified CM2 -> CUAC operator -> CUAC (Client)	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCUACG015	Queue color displaying for broadcast call (Intra - Call park)	Verify whether color for broadcast call is displayed correctly after retrieving the parked call in CUAC client GUI page when call comes to CUAC operator from an Unified IP Phone registered in Unified CM	Phone A -> Unified CM -> CUAC operator -> CUAC (Client)	Passed	
UCJ10SSCUACG016	Queue color displaying for broadcast call(SIP Trunk - Call Park)	Verify whether color for broadcast call is displayed correctly in CUAC client GUI page after retrieving the parked call, when call comes to CUAC operator from an Unified IP Phone registered in Unified CM1 via SIP trunk	Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Phone B -> Unified CM2 -> CUAC operator -> CUAC (Client)	Passed	

Cisco UC Integration™ for Microsoft Lync

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SCUCIG001	Making call from CUCI-Lync in softphone mode by hitting enter from the docked window	Verify whether the call has made successfully from CUCI-Lync in soft phone mode to IP Phone by hitting enter in the docked window	CUCI-Lync -> Unified CM1 -> Phone A	Passed	
UCJ105SCUCIG002	Making call from CUCI-Lync in softphone mode via docked window by selecting the docked window using keyboard short cut keys	Verify whether the call has made successfully from CUCI-Lync in softphone mode to IP Phone via docked window by selecting the docked window using keyboard shortcut keys	CUCI-Lync -> Unified CM1 -> Phone A	Passed	
UCJ105SCUCIG003	Transfer call from CUCI-Lync in desk phone mode to IP Phone via docked window by selecting the docked window using keyboard short cut keys	Verify whether the call is transferred successfully from CUCI-Lync in desk phone mode to IP Phone via docked window by selecting the docked window using keyboard short cut keys	CUCI-Lync -> Unified CM1 -> Phone A -> Unified CM1 -> Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SCUCIG004	Making call from CUCI-Lync in softphone mode to IP Phone at the different cluster via docked window by selecting the docked window using keyboard short cut keys	Verify whether the call has made successfully from CUCI-Lync in softphone mode to IP Phone at the different cluster through SIP trunk via docked window by selecting the docked window using keyboard shortcut keys	CUCI-Lync -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Phone A	Passed	
UCJ105SCUCIG005	Make call from CUCI-Lync to IP phone via search drop down in CUCI-Lync using keyboard short cut keys	Verify whether the call has made successfully from CUCI-Lync to IP phone via search drop down in CUCI-Lync using keyboard short cut keys	CUCI-Lync -> Unified CM1 -> Phone A	Passed	
UCJ105SCUCIG015	Change user dialog when logging into CUCI-Lync using the second user present in the same cluster	Verify whether the change user dialog is displayed when logging into CUCI-Lync with second user present in the same cluster	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCUCIG016	Change user dialog when logging into CUCI-Lync using an another user present in the same cluster and making an intra cluster call	Verify whether the change user dialog is displayed when logging into CUCI-Lync using an another user present in the same cluster and making call to the IP Phone at the same cluster	CUCI-Lync -> Unified CM1 -> Phone A	Passed	
UCJ10SSCUCIG017	Change user dialog when logging into CUCI-Lync using a second user present in the same cluster and making call to IP Phone A and transferring the call to IP Phone B at the different cluster	Verify whether the change user dialog is displayed when second user present in the same cluster is logging into CUCI-Lync and making call to IP Phone A and transferring the call to IP Phone B at the different cluster via SIP trunk	CUCI-Lync -> Unified CM1 -> IP Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2-> IP Phone B	Passed	
UCJ10SSCUCIG018	Change user dialog when logging into CUCI-Lync using an another user present in the same cluster and making an inter cluster call	Verify whether the change user dialog is displayed when logging into CUCI-Lync using an another user in the same cluster and making call to the IP Phone present at the different cluster via SIP trunk	CUCI-Lync -> Unified CM1 -> SIP Trunk -> Unified CM2 -> Phone A	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSCUCIG019	Change user dialog when logging into CUCI-Lync using a second user present in the same cluster and making an intra cluster call conference with IP Phones	Verify whether the change user dialog is displayed when logging into CUCI-Lync with second user present in the same cluster and making conference call to the IP Phones at the same cluster	CUCI-Lync -> Unified CM1 -> Phone A -> Unified CM1 -> Phone B	Passed	
UCJ10SSCUCIG035	Calling from CUCI-Lync in softphone mode to IP Phones via URI dialing	Verify whether the call made from CUCI-Lync in softphone mode to IP Phones via URI dialing is successful	CUCI-Lync -> Unified CM1 -> Phone A	Passed	
UCJ10SSCUCIG036	Call Transfer from CUCI-Lync in softphone mode to IP Phones via URI dialing	Verify whether the call transfer from CUCI-Lync in softphone mode to IP Phones via URI dialing is successful	Phone A -> Unified CM1 -> CUCI-Lync -> Unified CM1 -> Phone B	Passed	

Cisco TelePresence Multipoint Control Unit

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SMCUG001	Long duration Adhoc conference call using MCU 4510	Verify whether the Cisco TelePresence SX20 Quick Set ,Cisco TelePresence System 500-32 and Cisco Unified IP Phone 8945 registered in Cisco unified Communications Manager can join the conference call by Cisco TelePresence MCU 4510 and stay in the conference for 1 hour	8945 ->EX90>500-32->SX20 QuickSet ->Unified CM-> SIP Trunk->MCU 4510	Passed	
UCJ105SMCUG002	Hold and resume the adhoc conference call using MCU 4510	Verify whether the Cisco TelePresence SX20 Quick Set ,Cisco TelePresence system 500-32 and Cisco Unified IP Phone 8945 registered in Cisco unified Communications Manager can join the conference call by Cisco TelePresence MCU 4510 and hold and resume the conference call	8945 ->EX90>500-32->SX20 QuickSet -> Unified CM-> SIP Trunk->MCU 4510->Hold/Resume	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SMCUG003	Sharing the presentation in adhoc conference call using MCU 4510	Verify whether the Cisco TelePresence SX20 Quick Set ,Cisco TelePresence system 500-32 and Cisco Unified IP Phone 8945 registered in Cisco unified Communications Manager can join the conference call by Cisco TelePresence MCU 4510 and start presentation sharing in 500-32	8945 ->EX90->500-32->SX20 QuickSet ->Unified CM-> SIP Trunk->MCU 4510->Presentation Sharing	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SMCUG004	Annotation in adhoc conference call using MCU 4510	Verify whether the Cisco TelePresence SX20 Quick Set ,Cisco TelePresence system 500-32 and Cisco Unified IP Phone 8945 registered in Cisco unified Communications Manager can join the conference call by Cisco TelePresence MCU 4510 and start presentation sharing in 500-32 and annotate the Presentation	8945 ->EX90->500-32->SX20 QuickSet ->Unified CM->SIP Trunk-> MCU 4510->Presentation Sharing annotation	Passed	
UCJ105SMCUG005	Adhoc conference call between 8945 and EX60 login with EM	Verify that user is able to do ahoc conference call between 8945 and EX60 login with EM	8945(1)->EX60(EM)-8945(2) ->Unified CM->SIP Trunk->MCU 4510->Conference	Passed	
UCJ105SMCUG006	Hold and resume the adhoc conference call between 8945 and EX60 login with EM	Verify that user is able to do Hold and Resume the adhoc conference call between 8945 and EX60 login with EM	8945(1)->EX60(EM)->8945(2) ->Unified CM->SIP Trunk->MCU 4510->Conference ->Hold/Resume	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105SMCUG007	Presentation sharing in adhoc conference call between 8945 and EX60 login with EM using MCU 4510	Verify that user is able to do presentation sharing in adhoc conference call between 8945 and EX60 login with EM using MCU 4510	8945(1)->Unified CM->EX60(EM)->(add participant)->8945(2) ->conference ->Presentation sharing	Passed	
UCJ105SMCUG008	Transferring the adhoc conference call from EX60 login with EM to 500-32	Verify that user is able to transfer adhoc conference call from EX60 login with EM to 500-32	8945(1)-> Unified CM-> EX60 (EM) -> (add participant)-> 8945(2) EX60 (EM) -> Unified CM -> Transfer -> 500-32	Passed	
UCJ105SMCUG009	Presentation Sharing in 500-32 after transferring the adhoc conference call	Verify that user is able to do presentation sharing in 500-32 after transferring the adhoc conference call	8945(1) -> Unified CM->EX60(EM)->(add participant)->8945(2) EX60 (EM)->Unified CM -> Transfer -> 500-32 -> Presentation sharing	Passed	
UCJ105SMCUG010	Parking the adhoc conference call in 8945 using MCU 4510	Verify that user is able to park the adhoc conference call in 8945 using MCU 4510	8945(1) -> EX60 -> 8945(2)-> Unified CM-> SIP Trunk -> MCU 4510 -> Conference 8945(1) -> Unified CM-> Park the call	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSMCUG011	Adhoc conference in EX60 when it is in shared line with Codec C90	Verify that user is able to do adhoc conference in EX60 when it is in shared line with Codec C90	8945(1)->Unified CM->EX60 and Codec C90(Shared line) 8945(1)->Unified CM->8945(2) and EX60->Conference	Passed	
UCJ10SSMCUG012	Hold and Resume in EX60 when it is in shared line with Codec C90	Verify that user is able to Hold and Resume the adhoc conference in EX60 when it is in shared line with Codec C90	8945(1)->Unified CM->EX60 and Codec C90(Shared line) 8945(1)->Unified CM ->8945(2) and EX60-> Conference EX60(hold)->Unified CM-> Codec C90(Resume)	Passed	
UCJ10SSMCUG013	Presentation Sharing in EX60 when it is in shared line with EX90	Verify that user is able to do Presentation sharing during adhoc conference in EX60 when it is in shared line with EX90	8945(1) -> Unified CM -> EX60 and Codec C90(Shared line) 8945(1) -> Unified CM -> 8945(2) and EX60-> Conference-> Presentation Sharing	Passed	

Cisco TelePresence Management Suite

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SS.TMS.G.001	Schedule a conference call to endpoints registered in Unified CM using Conductor in Cisco TMS	Verify whether conference can be scheduled to endpoints registered in Cisco Unified Communications Manager using Cisco TelePresence Conductor which uses Cisco TelePresence MCU as media resource in Cisco TelePresence Management Suite	Cisco TMS->TelePresence Conductor->MCU 4510->SIP Trunk->Unified CM-> EX90, EX60, SX20 Quick Set	Passed	
UCJ10SS.TMS.G.002	Schedule a conference call to endpoints registered in Cisco VCS using Conductor in Cisco TMS	Verify whether conference can be scheduled to endpoints registered in Cisco TelePresence Video Communication Server using Cisco TelePresence Conductor which uses Cisco TelePresence MCU as media resource in Cisco TelePresence Management Suite	Cisco TMS->TelePresence Conductor->MCU 4510->Cisco VCS->EX90, EX60, SX20 Quick Set	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5.STMSG.003	Cisco TMS Phonebooks for EX90 registered in Unified CM	Verify whether the Cisco TelePresence Management Suite phonebook can be used in Cisco TelePresence System EX90 registered to Cisco Unified Communications Manager	NA	Passed	
UCJ10.5.STMSG.004	Recurrence information expanded in the booking confirmation email	Verify whether the conference recurrence information is sent via the booking confirmation email to all the participants in the conference in Cisco TelePresence management Suite	NA	Passed	
UCJ10.5.STMSG.005	Past meeting should not be deleted in Cisco TMS	Verify whether the meeting in the past can no longer be deleted in Cisco TelePresence Management Suite	NA	Passed	
UCJ10.5.STMSG.006	Ongoing meeting should not be deleted in Cisco TMS	Verify whether the ongoing meeting can no longer be deleted permanently in Cisco TelePresence Management Suite	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105S:TMSG.007	Extending the conference without any resource consideration in MCU 4510	Verify whether the conference can be extended when there is no resource consideration in the conference hosted by Cisco TelePresence MCU 4510 in Cisco TelePresence Management Suite	Cisco TMS->MCU 4510->SIP Trunk->Unified CM->EX90, EX60, SX20 Quick Set	Passed	
UCJ105S:TMSG.008	Extending the conference without any resource consideration in Conductor which uses MCU 4510 as conference bridge	Verify whether the conference can be extended when there is no resource consideration in the conference hosted by Cisco TelePresence Conductor which uses Cisco TelePresence MCU 4510 as Conference Bridge in Cisco TelePresence Management Suite	Cisco TMS -> Conductor -> MCU 4510 -> SIP Trunk -> Unified CM -> EX90, EX60, SX20 Quick Set	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5.TMS.G.009	Extending the conference without any resource consideration in Conductor which uses MCU 5310 as conference bridge	Verify whether the conference can be extended when there is no resource consideration in the conference hosted by Cisco TelePresence Conductor which uses Cisco TelePresence MCU 5310 as conference bridge in Cisco TelePresence Management Suite	Cisco TMS->Conductor->MCU 5310->SIP Trunk->Unified CM-> EX90, EX60, SX20 Quick Set	Passed	
UCJ10.5.TMS.G.010	Automatic extension for a conference using MCU 4510 as conference bridge	Verify whether the conference can be auto extended in the conference hosted by Cisco TelePresence MCU 4510 in Cisco TelePresence Management Suite	Cisco TMS-> MCU 4510->SIP Trunk->Unified CM->EX90, EX60, SX20 Quick Set	Passed	
UCJ10.5.TMS.G.011	Automatic Extension for a conference in Conductor which uses MCU 4510 as conference bridge	Verify whether the conference can be auto extended in the conference hosted by Cisco TelePresence Conductor which uses Cisco TelePresence MCU 4510 in Cisco TelePresence Management Suite	Cisco TMS->Conductor-> MCU 4510->SIP Trunk->Unified CM->EX90, EX60, SX20 Quick Set	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSTMSG012	Allocated Participants to join a conference before 5 minutes	Verify whether the participants who are added to the scheduled conference are added to the conference 5 minutes before the actual start time of the conference in Cisco TMS	NA	Passed	
UCJ10SSTMSG013	Replacing an EX90 registered with Unified CM by another system	Verify whether the Cisco TelePresence System EX90(1) registered to Cisco Unified Communications Manager can be replaced with another Cisco TelePresence System EX90 (2) registered with Cisco Unified Communications Manager.	NA	Passed	
UCJ10SSTMSG014	Replacing an EX90 registered with Unified CM by another system registered in Cisco VCS	Verify whether the Cisco TelePresence System EX90 (1) registered to Cisco Unified Communications Manager can be replaced with another Cisco TelePresence System EX90 (2) registered with Cisco TelePresence Video Communication Server	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSTMSG015	Delete option keys from Cisco TMS	Verify whether the option keys can be directly deleted from the Cisco TelePresence Management Suite in general setting page	NA	Passed	
UCJ10SSTMSG016	Cisco TelePresence Server 7010 Content Port information on Cisco TMS	Verify whether the Cisco TelePresence Server 7010 Content Port information can be viewed on Cisco TelePresence management Suite	NA	Passed	
UCJ10SSTMSG017	Schedule a conference call using Cisco TS 7010 SIP Trunked to Unified CM in Cisco TMS	Verify whether the scheduled conference can be established using Cisco TelePresence Server 7010 which is SIP Trunked to Cisco Unified Communications Manager	Cisco TMS->Cisco TS 7010->SIP Trunk->Unified CM->EX90, EX60, SX20 Quick Set	Passed	
UCJ10SSTMSG018	Conference email notification to participants	Verify whether the conference email notification to the participants can be sent using the smart scheduler in Cisco TelePresence Management Suite Provisioning Extension	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105S:TMSG:019	Mute on join video and audio for a schedule conference using Conductor in Cisco TMS for endpoints registered in Unified CM	Verify whether mute audio and video on join for scheduled conference works successfully to endpoints registered in Cisco Unified Communications Manager using Cisco TelePresence Conductor which uses Cisco TelePresence MCU as media resource in Cisco TelePresence Management Suite	Cisco TMS -> Mute on Join -> Conductor -> MCU 4510 -> SIP Trunk -> Unified CM-> EX90-> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5.STMS.G.020	Mute on join video and audio for a schedule conference using Conductor in Cisco TMS for endpoints registered in Cisco VCS	Verify whether mute audio and video on join for scheduled conference works successfully for endpoints registered in Cisco TelePresence Video Communication Server using Cisco TelePresence Conductor which uses Cisco TelePresence MCU as media resource in Cisco TelePresence Management Suite	Cisco TMS -> Mute on Join -> Conductor-> MCU 4510 -> Cisco VCS -> EX90 -> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105S:TMSG021	Initiating a scheduled audio conference using Conductor in Cisco TMS for endpoints registered in Unified CM	Verify whether the scheduled audio conference for Cisco TelePresence System EX90 and Cisco TelePresence System EX60 registered in Cisco Unified Communications Manager can be scheduled successfully using Cisco TelePresence Conductor which uses MCU 4510 as media resource in Cisco TelePresence Management Suite	Cisco TMS -> Audio -> Conductor ->MCU 4510 -> SIP Trunk -> Unified CM -> EX90 -> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSTMSG022	Initiating a scheduled audio conference using Conductor in Cisco TMS for endpoints registered in Cisco VCS	Verify whether the scheduled audio conference for Cisco TelePresence System EX90 and Cisco TelePresence System EX60 registered in Cisco Video Communication Server can be scheduled successfully using Cisco TelePresence Conductor which uses MCU 4510 as media resource in Cisco TelePresence Management Suite	Cisco TMS -> Audio -> Conductor -> MCU 4510 -> Cisco VCS -> EX90 -> EX60	Passed	
UCJ10SSTMSG023	Adding participants to the conference using Conference Control Center	Verify whether the participants can be added in the scheduled conference managed by Cisco TelePresence Conductor which uses MCU 4510 as media resource using Conference Control Center in Cisco TelePresence Management Suite	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ105S:TMSG.024	Conference using Collaboration Meeting Room which uses Cisco MCU 4510 as conference resource in Cisco TMS	Verify whether the conference can be created in the Collaboration Meeting Room in Cisco TelePresence Management Suite self care portal which uses Cisco TelePresence MCU 4510 as conference bridge	Collaboration Meeting Room -> Cisco TMS -> Conductor -> MCU 4510 -> Participants	Passed	
UCJ105S:TMSG.025	Conference using Collaboration Meeting Room which uses MCU 5310 as conference resource in Cisco TMS	Verify whether the conference can be created in the Collaboration Meeting Room in Cisco TelePresence Management Suite self care portal which uses Cisco TelePresence MCU 5310 as conference bridge	Collaboration Meeting Room -> Cisco TMS -> Conductor -> MCU 5310 -> Participants	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SSTMSG026	Audio conference using Collaboration Meeting Room which uses Cisco MCU 4510 as conference resource in Cisco TMS	Verify whether the audio conference can be created in the Collaboration Meeting Room in Cisco TelePresence Management Suite self care portal which uses Cisco TelePresence MCU 4510 as conference bridge	Collaboration Meeting Room -> Cisco TMS -> Conductor -> MCU 4510 -> Participants -> audio conference	Passed	
UCJ10SSTMSG027	Changing layout of a conference using Collaboration Meeting Room	Verify whether the layout can be changed in the Collaboration Meeting Room in Cisco TelePresence Management Suite self care portal which uses Cisco TelePresence MCU 4510 as conference bridge	Collaboration Meeting Room -> Cisco TMS -> Conductor -> MCU 4510 -> Participants-> Change Layout	Passed	

Cisco TelePresence Conductor

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TCG.001	Capacity of the Cisco TelePresence Server on VM conference bridge using New Capacity Management API	Verify whether the Capacity Management API returns the capacity of the conference bridge that will be used for the conference after giving the alias associated with Cisco TelePresence Server on VM.	NA	Passed	
UCJ10.5S.TCG.002	Capacity of the MCU 4510 conference bridge using New Capacity Management API	Verify whether the Capacity Management API returns the capacity of the conference bridge that will be used for the conference after giving the alias associated with Cisco TelePresence MCU 4510.	NA	Passed	
UCJ10.5S.TCG.003	Configure conference on Conductor using New Provisioning API	Verify whether the conference can be configured using the Provisioning API in Cisco TelePresence Conductor.	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TC.G.004	Alias Lookup in Conductor	Verify whether the lookup time for a conference with multiple alias configured is reduced in Cisco TelePresence Conductor.	NA	Passed	
UCJ10.5S.TC.G.005	Numeric dialing in Conductor which uses MCU 4510 as conference bridge	Verify whether the numeric dial string can be used while initiating a conference in Cisco TelePresence Conductor which uses Cisco MCU 4510 as conference bridge	EX90 -> Cisco VCS -> Conductor -> MCU 4510	Passed	
UCJ10.5S.TC.G.006	Same Conference display name for multiple conferences	Verify whether two or more conferences can be given the same conference display name in Cisco TelePresence Conductor.	NA	Passed	
UCJ10.5S.TC.G.007	Case Insensitive URI-Dialing	Verify whether SIP URI in upper case is converted to lowercase and the conference is established successfully in Cisco TelePresence Conductor	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TC.G.008	Collaboration Meeting Room information in the Web-GUI	Verify whether the Collaboration Meeting Room information is displayed in the Cisco TelePresence Conductor Web-GUI	NA	Passed	
UCJ10.5S.TC.G.009	Audio only adhoc conference using Conductor	Verify whether the participant quality can be set to audio only while using Cisco TelePresence Server as the conference bridge for adhoc conferences in Cisco TelePresence Conductor for endpoints registered in Cisco Unified Communications Manager	EX90 -> Unified CM-> EX60 -> conference -> Unified CM -> MRGL -> Conductor -> Cisco TS 7010 -> SX20 Quick Set -> adhoc conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TC.G.010	Audio only rendezvous conference using Conductor	Verify whether the participant quality can be set to audio only while using Cisco TelePresence Server as the conference bridge for rendezvous conferences in Cisco TelePresence Conductor for endpoints registered in Cisco Unified Communications Manager	EX90 -> EX60 -> SX20 Quick Set -> Unified CM -> SIP Trunk-> Conductor -> Cisco TS 7010 -> Rendezvous conference	Passed	
UCJ10.5S.TC.G.011	Numeric dialing in Conductor which uses Cisco TS 7010 as conference bridge	Verify whether the numeric dial string can be used while initiating a conference in Cisco TelePresence Conductor which uses Cisco TS 7010 as conference bridge	EX90 -> Cisco VCS -> Conductor -> Cisco TS 7010	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TC.G.012	Numeric dialing in Conductor which uses Cisco TelePresence Server on VM	Verify whether the numeric dial string can be used while initiating a conference in Cisco TelePresence Conductor which uses Cisco TelePresence Server on VM as conference bridge	EX90 -> Cisco VCS-> Conductor -> Cisco TelePresence Server on VM	Passed	
UCJ10.5S.TC.G.013	Audio only adhoc conference for Cisco TS 7010 using Conductor	Verify whether the participant quality can be set to audio only while using Cisco TelePresence Server 7010 as the conference bridge for adhoc conferences in Cisco TelePresence Conductor for endpoints registered in Cisco Unified Communications Manager	EX90 -> Unified CM-> EX60-> add participant-> Unified CM-> MRGL-> Conductor -> SX20 Quick Set -> adhoc conference	Passed	
UCJ10.5S.TC.G.014	Participant list of adhoc conference managed by Cisco TS 7010 in Conductor	Verify whether the participant list can be seen in Cisco TelePresence Conductor during an adhoc conference managed by Cisco Telepresence Server 7010	EX90 -> Unified CM -> EX60 -> add participant-> Unified CM -> MRGL-> Conductor -> SX20 Quick Set -> adhoc conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TC.G.015	Participant list of rendezvous conference managed by Cisco TS 7010 in Conductor	Verify whether the participant list can be seen in Cisco TelePresence Conductor during an rendezvous conference managed by Cisco Telepresence Server 7010	EX90 -> EX60-> SX20 Quick Set -> Unified CM-> SIP Trunk-> Conductor -> Cisco TelePresence Server 7010 -> Rendezvous conference	Passed	

Cisco TelePresence Server

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TS.G.001	Least recent speaker dropped down and newly joined speaker PiP position for Unified CM registered endpoints	To verify whether the least recent speaker is dropped down from the PiP strip and newly joined speaker is added to the PiP position for the endpoints registered with Unified CM	EX60, EX90 , Integrator Package C90 and 500-32->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.002	Least recent speaker dropped down and newly joined speaker PiP position for Cisco VCS registered endpoints	To verify whether the least recent speaker is dropped down from the PiP strip and newly joined speaker is added to the PiP position for the SIP endpoints registered with Cisco VCS	EX60, EX90 , Integrator Package C90 ->Cisco VCS->SIP trunk->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TS.G.003	Reappearance of hidden least recent speaker in the PiP for Unified CM registered endpoints	To verify whether the least recent speaker who was hidden from the PiP reappears in the PiP once that hidden participant starts speaking again for Unified CM registered endpoints	EX60, EX90, Integrator Package C90 and 500-32->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.004	Reappearance of hidden least recent speaker in the PiP for Cisco VCS SIP endpoints	To verify whether the least recent speaker who was hidden from the PiP reappears in the PiP once that hidden participant starts speaking again for Cisco VCS registered SIP endpoints	EX60, EX90, Integrator Package C90 ->Cisco VCS->SIP trunk->unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.005	Video announce for Unified CM registered endpoints while using ActivePresence layout	To verify whether the active speaker and participants are shown in PiP strip when they join to or drop from the conference for the endpoints registered with Unified CM	EX60, EX90 , Integrator Package C90 and 500-32->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TS.G.006	Video announce for Cisco VCS registered SIP endpoints while using ActivePresence layout	To verify whether the active speaker and participants are shown in PiP strip when they join to or drop from the conference for the SIP endpoints registered with Cisco VCS	EX60, EX90 , Integrator Package C90 ->Cisco VCS->SIP trunk->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.007	Video announce for Unified CM registered endpoints while using Equal layout	To verify whether the active speaker and participants are shown in PiP strip when they join to or drop from the conference for the endpoints registered with Unified CM	EX60, EX90 , Integrator Package C90 and 500-32->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.008	Mute/ Unmute audio using *6 combination on a DTMF keypad for Unified CM registered endpoints in conference	To verify whether the user is able to Mute/Unmute audio using *6 combination on a DTMF keypad for unified CM registered endpoints in a video conference	EX60, EX90, Integrator Package C90 ->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TS.G.009	Mute/ Unmute audio using *6 combination on a DTMF keypad for Cisco VCS SIP endpoints in conference	To verify whether the user is able to Mute/Unmute audio using *6 combination on a DTMF keypad for Cisco VCS registered SIP endpoints in a video conference	EX60, EX90, Integrator Package C90 ->Cisco VCS->SIP trunk->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.010	Mute/ Unmute audio using *6 combination on a DTMF keypad for Jabber for Iphone in conference	To verify whether the user is able to Mute/ Unmute audio using *6 combination on a DTMF keypad for unified CM registered Jabber for Iphone in a video conference	EX60, Jabber on Iphone, Integrator Package C90 ->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.011	Appearance of Mute Icon while using that endpoints own mute control for Unified CM registered endpoints	To verify whether the mute icon is appearing or not while using the endpoint's own mute control for unified CM registered endpoints	EX60, EX90, Integrator Package C90 ->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.012	Appearance of Mute Icon while using that endpoints own mute control for Cisco VCS registered endpoints	To verify whether the mute icon is appearing or not while using the endpoint's own mute control for Cisco VCS registered endpoints	EX60, EX90, Integrator Package C90 ->Unified CM->SIP trunk->Cisco TS 7010->video conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TS.G.013	Active speaker indicator and display(names on) for Unified CM registered endpoints in a conference	To verify whether the user is able to check active speaker indicator and display (names on) for Unified CM registered endpoints in a video conference	EX60, 500-32, Integrator Package C90 ->Unified CM->SIP trunk->Cisco VCS->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.014	Active speaker indicator and display(names on) for Cisco VCS SIP endpoints in a conference	To verify whether the user is able to check active speaker indicator and display (names on) for Cisco VCS SIP registered endpoints in a video conference	EX60, EX90, Integrator Package C90 ->Cisco VCS->SIP Trunk->Unified CM->SIP Trunk ->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.015	Active speaker indicator and display(names off) for Unified CM registered endpoints in a conference	To verify whether the user is able to check active speaker indicator and display (names off) for Unified CM registered endpoints in a video conference	EX60, 500-32, Integrator Package C90 ->Unified CM->SIP trunk->Cisco VCS->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.016	Active speaker indicator and display(names off) for Cisco VCS SIP endpoints in a conference	To verify whether the user is able to check active speaker indicator and display (names off) for Cisco VCS SIP registered endpoints in a video conference	EX60, EX90, Integrator Package C90 ->Cisco VCS->SIP Trunk->Unified CM->SIP Trunk ->Cisco TS 7010->video conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TS.G.017	Active speaker indicator and display(names off) for Unified CM registered endpoints in a conference using Cisco TS 7010 as a SIP Registrar	To verify whether the user is able to check active speaker indicator and display (names off) for Unified CM registered endpoints in a video conference using Cisco TelePresence Server 7010 as SIP registrar	EX60, 500-32, Integrator Package C90 ->Unified CM->SIP trunk->Cisco VCS->Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.018	Active speaker indicator and display(names off) for Cisco VCS SIP endpoints in a conference using Cisco TS 7010 as SIP registrar	To verify whether the user is able to check active speaker indicator and display (names off) for Cisco VCS SIP registered endpoints in a video conference using Cisco TelePresence Server 7010 as a SIP Registrar	EX60, EX90, Integrator Package C90 ->Cisco VCS-> Cisco TS 7010->video conference	Passed	
UCJ10.5S.TS.G.019	Notification banner for Unexpected media resource restart in Cisco TS 7010 when it is in locally managed mode	To verify whether the user is able to get notification banner for unexpected media resource restart in Cisco TelePresence Server 7010 when it is in locally managed mode	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TS.G.020	Notification banner for Unexpected device restart in Cisco TS 7010 when it is in locally managed mode	To verify whether the user is able to get notification banner for unexpected device restart in Cisco TelePresence Server 7010 when it is in locally managed mode	NA	Passed	
UCJ10.5S.TS.G.021	Initiating the conference from 500-32 and video endpoints join the same conference using Cisco TS 7010	To verify whether the user is able to initiate the video conference from 500-32 and video endpoints join the same conference without adding any third party interop key using Cisco TelePresence Server 7010	500-32 ,EX60 and Integrator Package C90->Unified CM->SIP Trunk->Cisco TS 7010->Video conference	Passed	
UCJ10.5S.TS.G.022	Initiating the video conference from TX9000 and video endpoints join the same conference using Cisco TS 7010	To verify whether the user is able to initiate the video conference from TX9000 multi-screen endpoint and video endpoints join the same conference without adding any third party interop key using Cisco TelePresence Server 7010	TX9000, EX60 and EX90->Unified CM->SIP Trunk->Cisco TS 7010->Video conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.TS.G.023	Enabling the console password for Cisco TS 7010	To verify whether the user is able to enable the console password for Cisco TS 7010 using Web GUI	NA	Passed	
UCJ10.5S.TS.G.024	Enabling the console password for Cisco TelePresence Server on VM	To verify that user is able to enable the console password for Cisco TelePresence Server on VM using Web GUI	NA	Passed	
UCJ10.5S.TS.G.025	Checking the Cisco TelePresence Server on VM network connection statistics via Web Interface	To verify that user is able to check the Cisco TelePresence Server on VM network connection statistics via Web Interface	NA	Passed	

Cisco Collaboration Expressway

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.001	Call forward all to SIP URI in SX20 Quick Set registered via Collaboration Edge	Verify whether the call forward all can be set to a SIP URI number of Cisco TelePresence System EX90 in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge	EX60->Unified CM->Cisco VCS-C->Cisco VCS-E-> SX20 Quick Set->call forward all(SIP URI)->Cisco VCS-E->Cisco VCS-C->Unified CM->EX90	Passed	
UCJ10.5S.CE.G.002	Call forward busy to SIP URI in SX20 Quick Set registered via Collaboration Edge	Verify whether the call forward busy can be set to a SIP URI number of Cisco TelePresence System EX90 in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge	SX20 Quick Set->Cisco VCS-E->Cisco VCS-C->Unified CM->8945 EX60->Unified CM->Cisco VCS-C->Cisco VCS-E-> SX20 Quick Set ->call forward busy(SIP URI)->Cisco VCS-E->Cisco VCS-C->Unified CM->EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.003	Call forward on No Answer Internal to SIP URI in SX20 Quick Set registered via Collaboration Edge	Verify whether the call forward on No Answer Internal can be set to a SIP URI number of Cisco TelePresence System EX90 in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge	EX60->Unified CM-> Cisco VCS-C->Cisco VCS-E -> SX20 Quick Set->call forward all No Answer Internal (SIP URI) ->Cisco VCS-E-> Cisco VCS - C -> Unified CM->EX90	Passed	
UCJ10.5S.CE.G.004	Call forward on Unregistered to SIP URI in SX20 Quick Set registered via Collaboration Edge	Verify whether the call forward on Unregistered can be set to a SIP URI number of Cisco TelePresence System EX90 in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge	EX60->Unified CM->Cisco VCS-C->Cisco VCS-E->SX20 Quick Set ->call forward all Unregistered(SIP URI)->Cisco VCS-E->Cisco VCS-C->Unified CM->EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.005	Call forward all to EX90 registered in Cisco VCS in SX20 Quick Set registered via Collaboration Edge	Verify whether the call forward all can be set to Cisco TelePresence System EX90 registered in Cisco TelePresence Video Communication Server in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge	EX60->Cisco VCS->SIP Trunk->Unified CM->Cisco VCS-C->Cisco VCS-E->SX20 Quick Set->call forward all->Cisco VCS-E->Cisco VCS-C->Unified CM->SIP Trunk->Cisco VCS->EX90	Passed	
UCJ10.5S.CE.G.006	Call forward busy to EX90 registered in Cisco VCS in SX20 Quick Set Registered via Collaboration Edge	Verify whether the call forward busy can be set to Cisco TelePresence System EX90 registered in Cisco TelePresence Video Communication Server in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge	SX20 Quick Set -> Cisco VCS-C-> Cisco VCS-E -> Unified CM-> 8945 EX60 -> Unified CM -> Cisco VCS-C-> Cisco VCS-E-> SX20 Quick Set ->call forward busy -> Cisco VCS-E -> Cisco VCS-C ->Unified CM->SIP Trunk->Cisco VCS->EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.007	Call forward on Unregistered to EX90 registered in Cisco VCS in SX20 Quick Set registered via Collaboration Edge	Verify whether the call forward on Unregistered can be set to a Cisco TelePresence System EX90 registered in Cisco TelePresence Video Communications Server in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge	EX60->Unified CM->Cisco VCS-C->Cisco VCS-E->SX20 Quick Set ->call forward all Unregistered->Cisco VCS-E->Cisco VCS-C->Unified CM->SIP Trunk->Cisco VCS->EX90	Passed	
UCJ10.5S.CE.G.008	Join a TelePresence Conductor rendezvous conference in SX20 Quick Set	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge can join a Rendezvous conference managed by Cisco TelePresence Conductor	SX20 Quick Set -> Cisco VCS-E -> Cisco VCS-C -> Unified CM-> SIP Trunk -> Conductor-> TelePresence Server on VM	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.009	Long duration calls between SX20 Quick Set registered via Collaboration Edge and EX90	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge can make a call to Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager and stay in the call for 1 hour	SX20 Quick Set -> Cisco VCS-E->Cisco VCS-C->Unified CM -> EX90	Passed	
UCJ10.5S.CE.G.010	Long duration calls between SX20 Quick Set and EX90 both registered via Collaboration Edge	Verify whether the Cisco TelePresence SX20 Quick Set can make a call to Cisco TelePresence System EX90 both registered in Cisco Unified Communications Manager in the remote network via Collaboration Edge and stay in the call for 1 hour	SX20 Quick Set ->EX90->Cisco VCS-E->Cisco VCS-C->Unified CM	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.011	Long duration video calls between SX20 Quick Set registered via Collaboration Edge and EX90 registered in Cisco VCS	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager registered via Collaboration Edge can make a video call to Cisco TelePresence System EX90 registered in Cisco TelePresence Video Communication Server and stay in the call for 1 hour	SX20 Quick Set -> Cisco VCS-E->Cisco VCS-C->Unified CM->SIP Trunk ->Cisco VCS-> EX90	Passed	
UCJ10.5S.CE.G.012	Long duration audio calls between SX20 Quick Set registered via Collaboration Edge and 6961	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge can make a call to Cisco Unified IP phone 6961 registered in Cisco Unified Communications Manager and stay in the call for 1 hour	SX20 Quick Set -> Cisco VCS-E->Cisco VCS-C->Unified CM ->6961	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.013	Video call between SX20 Quick Set registered via Collaboration Edge and EM enabled EX90	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge can make a call to Extension Mobility enabled Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager	SX20 Quick Set -> Cisco VCS-E->Cisco VCS-C->Unified CM ->EM enabled EX90	Passed	
UCJ10.5S.CE.G.014	Consultative call transfer from SX20 Quick Set registered via Collaboration Edge to EM enabled EX90	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge can make a consultative transfer to Extension Mobility enabled Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager	SX20 Quick Set -> Cisco VCS-E->Cisco VCS-C->Unified CM -> EX60 SX20 Quick Set -> Hold / Transfer->Cisco VCS-E->Cisco VCS-C->Unified CM->EM enabled EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.015	Blind call transfer from SX20 Quick Set registered via Collaboration Edge to EM enabled EX90	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager via Collaboration Edge can make a consultative transfer to Extension Mobility enabled Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager	SX20 Quick Set -> Cisco VCS-E->Cisco VCS-C->Unified CM -> EX60 SX20 Quick Set-> Transfer->Cisco VCS-E->Cisco VCS-C->Unified CM->EM enabled EX90	Passed	
UCJ10.5S.CE.G.016	Initiating the Meet Me video conference from EX90 using Cisco TS 7010 via Collaboration Edge	To verify that user is able to initiate the meet me video conference from EX90 registered with Unified CM using Cisco TelePresence Server 7010 via Collaboration Edge	EX90-> Cisco VCS- E ->Cisco VCS - C ->Unified CM -> SIP Trunk ->Cisco TS 7010->Video Conference	Passed	
UCJ10.5S.CE.G.017	Video conference between video endpoints registered with unified CM using Cisco TS 7010 via Collaboration Edge	To verify that user is able to initiate the meet me video conference from EX90 registered with Unified CM using Cisco TelePresence Server 7010 via Collaboration Edge	EX90-> Cisco VCS- E ->Cisco VCS - C ->Unified CM->SIP Trunk-> Cisco TS 7010->Video Conference EX60 and C90->Unified CM->SIP Trunk->Cisco TS 7010->join the conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.018	Sharing the Presentation in Video conference using Cisco TS 7010 via Collaboration Edge	To verify that user is able to sharing the presentation in video conference using Cisco TS 7010 via Collaboration edge	EX90-> Cisco VCS- E ->Cisco VCS - C ->Unified CM->SIP Trunk-> Cisco TS 7010->Video Conference EX60 and C90->Unified CM->SIP Trunk->Cisco TS 7010->join the conference ->Presentation sharing	Passed	
UCJ10.5S.CE.G.019	Video conference between video endpoints and 8945 using Cisco TS 7010 via Collaboration Edge	To verify that user is able to do video conference between video endpoints and8945 IP Phone using Cisco TS 7010 via Collaboration Edge	EX90, EX60 and 8945 ->Cisco VCS - E-> Cisco VCS - C->Unified CM-> SIP Trunk->Cisco TS 7010->Video Conference	Passed	
UCJ10.5S.CE.G.020	Video conference between CTS 500-32 and video endpoints registered with Unified CM via Collaboration edge	To verify that user is able to do video conference between CTS 500-32 and video endpoints registered with unified CM via Collaboration edge	EX90->Cisco VCS - E-> Cisco VCS - C->>Unified CM-> SIP Trunk-> Cisco TS 7010->Video Conference EX60 and CTS 500-32->Unified CM->SIP Trunk->Cisco TS 7010->Video conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.021	Park and retrieve the video conference call via collaboration edge	To verify that user is able to park and retrieve the video conference call via collaboration edge	EX90 ->Cisco VCS - E-> Cisco VCS - C->Unified CM->SIP Trunk-> Cisco TS 7010->Video Conference EX60 and 9971->Unified CM-> SIP Trunk->Cisco TS 7010->Video conference 9971->unified CM ->park->SX20 Quickset->Retrieve the call	Passed	
UCJ10.5S.CE.G.022	Adhoc conference from EX90 via Collaboration Edge	To verify that user is able to do adhoc conference from EX90 via collaboration Edge	EX90 and EX60 ->Cisco VCS - E-> Cisco VCS - C->Unified CM ->9971-> Adhoc conference	Passed	
UCJ10.5S.CE.G.023	Park and retrieve the video conference call using 7861 IP Phone and EX90 via collaboration edge	To verify that user is able to park and retrieve the video conference call using 7861 IP Phone and EX90 via collaboration edge	EX90(1) and EX60 ->Cisco VCS - E-> Cisco VCS - C-> Unified CM->SIP Trunk-> Cisco TS 7010->Video Conference 7861->Cisco VCS - E-> Cisco VCS - C-> Unified CM -> Park-> EX90(2) -> Retrieve	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.024	Park and retrieve the video conference call using 8831 IP Phone and EX60 via collaboration edge	To verify that user is able to park and retrieve the video conference call using 8831 IP Phone and EX60 via collaboration edge	EX60(1) and EX60(2) -> Cisco VCS - E-> Cisco VCS - C-> Unified CM -> SIP Trunk -> Cisco TS 7010->Video Conference 8831->Cisco VCS - E-> Cisco VCS - C-> Unified CM -> Park-> EX90-> Retrieve	Passed	
UCJ10.5S.CE.G.025	Hold and Resume in video conference via collaboration edge	To verify that user is able to do hold and resume in video conference call via collaboration edge	EX90 and EX60 ->Cisco VCS - E-> Cisco VCS - C-> Unified CM-> SIP Trunk->Cisco TS 7010->Video Conference->Hold / Resume	Passed	
UCJ10.5S.CE.G.026	Changing the presentation layout in video conference via collaboration edge	To verify that user is able to changing the presentation layout in video conference call via collaboration edge	EX90 and EX60 ->Cisco VCS - E-> Cisco VCS - C->Unified CM-> SIP Trunk->Cisco TS 7010->Video Conference 8945->Unified CM->SIP Trunk->Cisco TS 7010->Video conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.027	Video conference between VCS registered endpoints and Unified CM registered endpoints via collaboration edge	To verify that user is able to do Video conference between VCS registered endpoints and Unified CM registered endpoints via collaboration edge	EX90 and EX60 ->Cisco VCS - E-> Cisco VCS - C->Unified CM->SIP Trunk->Cisco TS 7010->Video Conference SX20 Quickset->Cisco VCS-> Unified CM->SIP Trunk->Cisco TS 7010->Video conference	Passed	
UCJ10.5S.CE.G.028	Consultative call transfer via collaboration edge	To verify that user is able to do Consultative call transfer via collaboration edge	EX90 and EX60 ->Cisco VCS - E-> Cisco VCS - C-> Unified CM EX90 -> unified CM- > Hold / Transfer-> SX20 Quickset	Passed	
UCJ10.5S.CE.G.029	Mute/ Unmute audio using *6 combination on a DTMF keypad for Cisco VCS SIP endpoints in video conference via collaboration edge	To verify that user is able to Mute/ Unmute audio using *6 combination on a DTMF keypad for Cisco VCS SIP endpoints in video conference via collaboration edge	EX90 and EX60 ->Cisco VCS - E-> Cisco VCS - C-> Unified CM->SIP Trunk->Cisco TS 7010->Video Conference SX20 Quickset->Cisco VCS->Unified CM->SIP Trunk->Cisco TS 7010->Video conference->Press *6	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10.5S.CE.G.030	Mute/ Unmute audio using *6 combination on a DTMF keypad for Unified CM endpoints in video conference via collaboration edge	To Verify that user is able to Mute/ Unmute audio using *6 combination on a DTMF keypad for Unified CM endpoints in video conference via collaboration edge	EX90 and EX60 ->Cisco VCS - E-> Cisco VCS - C-> Unified CM->SIP Trunk->Cisco TS 7010->Video Conference->Press *6 on EX90	Passed	

Regression

Regression Execution							
Features Tested	**Total test cases**	**Passed**	**Passed %**	**Passed W/X**	**Passed W/X %**	**Failed**	**Failed %**
Basic Call	19	19	100%	0	0%	0	0%
Call Park	12	12	100%	0	0%	0	0%
Call Transfer	19	19	100%	0	0%	0	0%
Blind Transfer	4	4	100%	0	0%	0	0%
Consult Transfer	3	3	100%	0	0%	0	0%
Conference	26	26	100%	0	0%	0	0%
Conference List	7	7	100%	0	0%	0	0%
Call Forward All	19	19	100%	0	0%	0	0%
Call Forward Busy	5	5	100%	0	0%	0	0%
Call Pickup	16	16	100%	0	0%	0	0%
Hold	2	2	100%	0	0%	0	0%
Hold and Resume	9	9	100%	0	0%	0	0%
Total	141	141	100%	0	0%	0	0%

Related Documentation

Cisco Unified Communications Manager

Release Note:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/rel_notes/10_5_1/CUCM_BK_CE15D2A0_00_cucm-release-notes-1051/CUCM_BK_CE15D2A0_00_cucm-release-notes-1051_chapter_01.html

Documentation Guide:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/docguide/10_5_1/CUCM_BK_C2CFB1F8_00_cucm-documentation-guide-1051.html

Cisco Unified Communications Self Care Portal User Guide 10.5.1

User Guide - Self-Care Portal:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/useroptions/10_5_1/CUCM_BK_U437D0F8_00_self-care-user-guide-1051.html

Cisco Unified IP Phones

Cisco IP Phone 78xx Release Note:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cuipph/7821_7841_7861/10_1/english/release_notes/PA2D_BK_R1B754C7_00_rn-10_1_1-7821-41-61-sr1/PA2D_BK_R1B754C7_00_rn-10_1_1-7821-41-64-sr1_chapter_00.html

Cisco Unified IP Phone 894x Release Note:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cuipph/8941_8945/firmware/9_4_1SR1/english/P415_BK_R69D0657_00_release-notes-941sr1/P415_BK_R69D0657_00_release-notes-941sr1_chapter_00.html

Cisco Unified Wireless IP Phone 792x Release Note

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cuipph/7925g_7925gEX_7926/firmware/1-4-5SR1/english/release-notes/P256_BK_W50717FB_00_wireless-phone-release-notes-1_4_5sr1/P256_BK_W50717FB_00_wireless-phone-release-notes-1_4_5sr1_chapter_00.html

Cisco Unified Survivable Remote Site Telephony:

Admin guide:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cusrst/admin/sccp_sip_srst/configuration/guide/SCCP_and_SIP_SRST_Admin_Guide.html

Cisco Unified Communications Manager IM and Presence

Admin guide:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/im_presence/configAdminGuide/10_5_1/CUP0_BK_CE43108E_00_config-admin-guide-imp-105.html

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cucm/im_presence/im_compliance/10_5_1/CUP0_BK_IFBE75ED_00_imp-im-compliance-guide-105.html

Cisco Unity Connection:

Release note:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/10x/release/notes/10_5_1cucrn.html

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/connection/vmo/release/notes/101cucvmorn.html

Cisco Jabber for Mac:

Installation and Configuration Guide:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/jabber/mac/9-6/JABM_BK_J40FF63B_00_jabber-mac-install-config-9-6.pdf

Release note:

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/jabber/mac/9-6/JABM_BK_J75BED49_00_jabber-mac-release-notes-9-6.pdf

Cisco VCS

Admin guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/vcs/admin_guide/Cisco-VCS-Administrator-Guide-X8-1-1.pdf

Release note:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/vcs/release_note/Cisco-VCS-Release-Note-X8-1-1.pdf

Configuration guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/vcs/config_guide/X8-1/Cisco-VCS-Basic-Configuration-Single-VCS-Control-Deployment-Guide-X8-1.pdf

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/vcs/config_guide/X8-1/Mobile-Remote-Access-via-VCS-Deployment-Guide-X8-1-1.pdf

Cisco Telepresence conductor

Admin guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/conductor/admin_guide/TelePresence-Conductor-Admin-Guide-XC2-3.pdf

Installation Guide

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/conductor/install_guide/TelePresence-Conductor-Virtual-Machine-Install-Guide-XC2-3.pdf

Release note:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/conductor/release_note/TelePresence-Conductor-Release-Notes-XC2-3.pdf

Cisco TMS

Admin guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/tms/admin_guide/Cisco-TMS-Admin-Guide-14-4.pdf

Installation Guide

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/tms/install_guide/Cisco-TMS-Install-Guide-14-4.pdf

Release note:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/tms/release_note/Cisco-TMS-release-notes-14-4.pdf

Cisco TelePresence Server 7010

Release note:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/ts/release_note/Cisco-TelePresence-Server-Software-Release-Notes-4-0-1-57.pdf

Installation Guide

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/ts/install_guide/Cisco_TelePresence_Server_7010_Installation_Guide.pdf

Deployment Guide

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/ts/deployment_guide/Cisco_TelePresence_Server_Deployment_Guide.pdf

Cisco TelePresence Server on VM

Installation Guide

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/ts/install_guide/Cisco-TelePresence-Server-on-Virtual-Machine-Install-Guide.pdf

Cisco TelePresence MCU:

Admin guide:

http://www.cisco.com/en/US/docs/telepresence/infrastructure/mcu/admin_guide/Cisco_TelePresence_MCU_4-4_Product_administration_guide.pdf

Release note:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/mcu/release_note/Cisco-TelePresence-MCU-Software-release-note-4-4-3-67.pdf

Cisco TelePresence TX9000

Release note:

http://www.cisco.com/c/en/us/td/docs/telepresence/tx_sw/6_0/release/notes/tx_sw_6_0_release_notes.html

Admin guide:

http://www.cisco.com/c/en/us/td/docs/telepresence/tx_sw/6_0/admin/guide/tx_6_0_admin_guide/tx_sw_features.html

Cisco TelePresence System Integrator package C90

Getting Started guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/codec-c-series/tc7/getting-started-guide/video-systems-getting-started-guide-tc70.pdf

Admin guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/codec-c-series/tc7/administration-guide/profile-c90-and-codec-c90-administrator-guide-tc70.pdf

Release note:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/software/tc7/release_notes/tc-software-release-notes-tc7.pdf

Cisco TelePresence System EX90

Getting Started guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/ex-series/tc7/getting-started-guide/video-systems-getting-started-guide-tc70.pdf

Admin guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/ex-series/tc7/administration-guide/ex-series-administrator-guide-tc70.pdf

Release note:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/software/tc7/release_notes/tc-software-release-notes-tc7.pdf

Cisco TelePresence SX20 Quickset

Getting Started guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/quick-set-sx20/tc7/getting-started-guide/video-systems-getting-started-guide-tc70.pdf

Admin guide:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/quick-set-sx20/tc7/administration-guide/sx20-quickset-administrator-guide-tc70.pdf

Release note:

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/endpoint/software/tc7/release_notes/tc-software-release-notes-tc7.pdf