



## DEFENSE INFORMATION SYSTEMS AGENCY

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IN REPLY  
REFER TO: Joint Interoperability Test Command (JTE)

**18 Jun 13**

### MEMORANDUM FOR DISTRIBUTION

**SUBJECT:** Extension of the Special Interoperability Test Certification of the Cisco Unified Communications Manager (CUCM) Version 8.6.1

References: (a) DoD Directive 4630.05, "Interoperability and Supportability of Information Technology (IT) and National Security Systems (NSS)," 5 May 2004  
(b) CJCSI 6212.01E, "Interoperability and Supportability of Information Technology and National Security Systems," 15 December 2008  
(c) through (f), see Enclosure

1. References (a) and (b) establish the Defense Information Systems Agency (DISA) Joint Interoperability Test Command (JITC), as the responsible organization for interoperability test certification.
2. The CUCM Version 8.6.1 is hereinafter referred to as the System Under Test (SUT). The SUT is certified for joint use in the Defense Information System Network (DISN) as a Local Session Controller (LSC). The fielding of the SUT is limited to Internet Protocol (IP) version 4 (IPv4) across the DISN. Although the SUT supports Internet Protocol version 6 (IPv6), it was not tested inter-enclave because of test limitations within the network infrastructure. Therefore, inter-enclave IPv6 is not covered under this certification, but intra-enclave use of IPv6 is authorized for use. Any new discrepancy noted in the operational environment will be evaluated for impact on the existing certification. These discrepancies will be adjudicated to the satisfaction of DISA via a vendor Plan of Action and Milestones (POA&M), which will address all new critical Test Discrepancy Reports (TDRs) within 120 days of identification. Testing was conducted using LSC product requirements derived from the Unified Capabilities Requirements (UCR), Reference (c), and LSC test procedures, Reference (d). No other configurations, features, or functions, except those cited within this memorandum, are certified by JITC. This certification expires upon changes that affect interoperability, but no later than the date of the original Unified Capabilities (UC) Approved Products List (APL) memorandum expiration (28 June 2015).
3. The extension of this certification is based upon Desktop Reviews (DTRs) 6 and 7. The original certification, documented in Reference (e), is based on interoperability testing conducted by JITC, review of the vendor's Letters of Compliance (LoC), DISA adjudication of open TDRs, and DISA Information Assurance (IA) Certification Authority (CA) approval of the IA configuration. Interoperability testing was conducted by JITC, Fort Huachuca, Arizona, from

11 July through 5 August 2011. Verification and Validation (V&V) testing was conducted by JITC, Fort Huachuca, Arizona, from 5 through 23 December 2011. Review of the vendor's LoC was completed on 19 June 2012. Adjudication of open TDRs was completed by DISA on 31 July 2012. The DISA CA provided a positive recommendation on 14 June 2012 based on the security testing completed by DISA-led IA test teams and published in a separate report, Reference (f). DTR 6 was requested to update the Integrated Services Router (ISR) second generation (G2) Interworking Gateways (IWGs) from 15.1(4)M2 to 15.2(4)M3. DTR 7 was requested to update the Voice Gateway (VG)350 from 15.2(4)M2 to 15.2(4)M3. The update for the IWG includes fixes that support the SUT's ability to support REFER messages with the Cisco EBC. JITC determined V&V testing was required for these DTRs. JITC conducted interoperability testing on 28 and 29 May 2013.

No new interoperability discrepancies were opened as a result of this V&V test. One previous interoperability discrepancy was closed. The VG350 gateway crashes when Precedence Call Waiting is invoked subsequently on an analog line. This only affects a precedence call above ROUTINE and is only caused when a second call above ROUTINE is placed to the same line in succession. This discrepancy was fixed and successfully tested with DTR 7, which included 15.2(4)M3 on the VG350. Additionally, the DISA CA has approved these DTRs to include 15.2(4)M3 without further testing. Therefore, the original IA approval applies to these DTRs. Therefore, JITC approves these DTRs.

4. The interface, Capability Requirements (CR) and Functional Requirements (FR), and component status of the SUT is listed in Tables 1 and 2. The threshold CR/FRs for LSCs are established by Sections 5.3.2, 5.3.4, 5.3.5, and 5.4 of Reference (c) and were used to evaluate the interoperability of the SUT.

**Table 1. SUT Interface Interoperability Status**

Interface	Critical	UCR Reference	Threshold CR/FR Requirements (See note 1.)	Status	Remarks
<b>Line Interfaces</b>					
10Base-X	Yes	5.3.2.6.3	2, 4, 10, 13, and 16	Certified	Met threshold CRs/FRs for IEEE 802.3i and 802.3j. Applies to PEIs and softphones.
100Base-X	Yes	5.3.2.6.3	2, 4, 10, 13, and 16	Certified	Met threshold CRs/FRs for IEEE 802.3u. Applies to PEIs and softphones.
1000Base-X	No	5.3.2.6.3	2, 4, 10,13, and 16	Certified	Met threshold CRs/FRs for IEEE 802.3ab. Applies to PEIs and softphones.
2-wire analog	Yes	5.3.2.6.1.6	2, 4, 10, and 13	Certified	Met threshold CRs/FRs for 2-wire instruments. Applies to 2-wire secure and non-secure analog instruments.
BRI	No	5.3.2.6.1.8	2, 4, 10, and 13	Not Tested	This interface is offered by the SUT; however, it was not tested because it does not support Assured Services.
<b>External Interfaces</b>					
10Base-X	No (See note 2.)	5.3.2.4.2	1, 2, 3, 6, 7, 8, 10, 11, 13, 15, and 16	Certified	Met threshold CRs/FRs for IEEE 802.3i and 802.3j. Applies to AS-SIP trunk.
100Base-X	No (See note 2.)	5.3.2.4.2	1, 2, 3, 6, 7, 8, 10, 11, 13, 15, and 16	Certified	Met threshold CRs/FRs for IEEE 802.3u. Applies to AS-SIP trunk.
1000Base-X	No (See note 2.)	5.3.2.4.2	1, 2, 3, 6, 7, 8, 10, 11, 13, 15, and 16	Certified	Met threshold CRs/FRs for IEEE 802.3z and 802.3ab. Applies to AS-SIP trunk.
ISDN T1 PRI ANSI T1.619a	Yes	5.3.2.4.3	2, 3, 7, 8, 10, and 13	Certified	Met threshold CRs/FRs. Provides legacy DSN and TELEPORT connectivity.

**Table 1. SUT Interface Interoperability Status (continued)**

Interface	Critical	UCR Reference	Threshold CR/FR Requirements (See note 1.)	Status	Remarks
<b>External Interfaces (continued)</b>					
ISDN T1 PRI NI-2	Yes	5.3.2.4.3	2, 3, 7, 8, 10, and 13	Certified	Met threshold CRs/FRs. Provides PSTN connectivity.
T1 CCS7 ANSI T1.619a	No	5.3.2.12.9	2, 3, 7, 8, 10, and 13	Not Tested	This interface is not offered by the SUT.
T1 CAS	No	5.3.2.12.11	2, 3, 7, 8, 10, and 13	Certified	Met threshold CRs/FRs for DTMF.
E1 CAS (DP, DTMF, MFR1)	No	5.3.2.12.11	2, 3, 7, 8, 10, 13	Not Tested	This interface is offered by the SUT; however, it was not tested and is not covered under this certification.
E1 PRI ITU-T Q.955.3	No (See note 3.)	5.3.2.12.10	2, 3, 7, 8, 10, and 13	Certified	This interface was tested and certified with Desktop Review 3.
E1 PRI ITU-T Q.931	No (See note 3.)	5.3.2.12.10	2, 3, 7, 8, 10, and 13	Certified	Met threshold CRs/FRs for European PSTN connectivity.
<b>NM</b>					
10Base-X	No (See note 2.)	5.3.2.4.4 5.3.2.7.2.8	16 and 17	Certified	Met threshold CRs/FRs. Verified via LoC.
100Base-X	No (See note 2.)	5.3.2.4.4 5.3.2.7.2.8	16 and 17	Certified	Met threshold CRs/FRs. Verified via LoC.
<b>NOTES:</b>					
1. The SUT high-level CR and FR ID numbers depicted in the Threshold CRs/FRs column can be cross-referenced in Table 2. These high-level CR/FR requirements refer to a detailed list of requirements provided in Reference (e), Enclosure 3.					
2. The SUT must provide a minimum of one of the listed interfaces.					
3. This interface is conditionally required for deployment in Europe.					
<b>LEGEND:</b>					
10Base-X	10 Mbps Ethernet		IEEE	Institute of Electrical and Electronics Engineers	
100Base-X	100 Mbps Ethernet		ISDN	Integrated Services Digital Network	
1000Base-X	1000 Mbps Ethernet		ITU-T	International Telecommunication Union – Telecommunication Standardization Sector	
802.3ab	1000 Mbps Ethernet over Twisted Pair		LoC	Letter of Compliance	
802.3i	10 Mbps twisted pair media for 10Base-X networks		Mbps	Megabits per second	
802.3j	10 Mbps fiber media for 10Base-X networks		MFR1	Multi-Frequency Recommendation 1	
802.3u	100BASE-TX, 100BASE-T4, 100BASE-FX Fast Ethernet at 100 Mbps with auto negotiation		MG	Media Gateway	
802.3z	Standard for Gigabit Ethernet		MLPP	Multi-Level Precedence and Preemption	
ANSI	American National Standards Institute		NI-2	National ISDN Standard 2	
AS-SIP	Assured Services Session Initiation Protocol		NM	Network Management	
BRI	Basic Rate Interface		PEI	Proprietary End Instrument	
CAS	Channel Associated Signaling		PRI	Primary Rate Interface	
CCS7	Common Channel Signaling		PSTN	Public Switched Telephone Network	
CR	Capability Requirement		Q.931	Signaling Standard for ISDN	
DP	Dial Pulse		Q.955.3	ISDN Signaling Standard for E1 MLPP	
DSN	Defense Switched Network		SS7	Signaling System 7	
DTMF	Dual Tone Multi-Frequency		SUT	System Under Test	
E1	European Basic Multiplex Rate (2.048 Mbps)		T1	Digital Transmission Link Level 1 (1.544 Mbps)	
FR	Functional Requirement		T1.619a	SS7 and ISDN MLPP Signaling Standard for T1	
ID	Identification		UCR	Unified Capabilities Requirements	

**Table 2. SUT Capability Requirements and Functional Requirements Status**

CR/FR ID	Capability/ Function	Applicability (See note 1.)	UCR Reference	Status
1	<b>Assured Services Product Features and Capabilities</b>			
	DSCP Packet Marking	Required	5.3.2.2.1.4	Met
	Voice Features and Capabilities	Required	5.3.2.2.2.1	Partially Met (See notes 2 and 3.)

**Table 2. SUT Capability Requirements and Functional Requirements Status (continued)**

CR/FR ID	Capability/ Function	Applicability (See note 1.)	UCR Reference	Status
1	<b>Assured Services Product Features and Capabilities (continued)</b>			
	Public Safety Features	Required	5.3.2.2.2.2	Met
	ASAC – Open Loop	Required	5.3.2.2.2.3	Met
	Signaling Protocols	Required	5.3.2.2.3	Met
	Signaling Performance	Conditional	5.3.2.2.4	Met
2	<b>Registration, Authentication, and Failover</b>			
	Registration	Required	5.3.2.3.1	Met
	Failover	Required	5.3.2.3.2	Met (See note 4.)
3	<b>Product Physical, Quality, and Environmental Factors</b>			
	Availability	Required	5.3.2.5.2.1	Met
	Maximum Downtimes	Required	5.3.2.5.2.2	Met
	Loss of Packets	Required	5.3.2.5.4	Met
4	<b>Voice End Instruments</b>			
	Tones and Announcements	Required	5.3.2.6.1.1	Met (See note 5.)
	Audio Codecs	Required	5.3.2.6.1.2	Met (See notes 5 and 6.)
	VoIP PEI or AEI Audio Performance	Required	5.3.2.6.1.3	Met (See note 5.)
	VoIP Sampling Standard	Required	5.3.2.6.1.4	Met (See note 5.)
	Authentication to LSC	Required	5.3.2.6.1.5	Met (See note 5.)
	Analog Telephone Support	Required	5.3.2.6.1.6	Met
	Softphones	Conditional	5.3.2.6.1.7	Partially Met (See note 7.)
	ISDN BRI	Conditional	5.3.2.6.1.8	Not Tested
5	<b>Video End Instruments</b>			
	Video End Instrument	Required	5.3.2.6.2	Met (See note 8.)
	Display Messages, Tones, and Announcements	Required	5.3.2.6.2.1	Met (See note 8.)
	Video Codecs (Including Associated Audio Codecs)	Required	5.3.2.6.2.2	Met (See note 8.)
6	<b>LSC Requirements</b>			
	PBAS/ASAC Requirements	Required	5.3.2.7.2.1	Met
	Calling Number Delivery Requirements	Required	5.3.2.7.2.2	Met
	LSC Signaling Requirements	Required	5.3.2.7.2.3	Met
	Service Requirements under Total Loss of WAN Transport	Required	5.3.2.7.2.4	Met
	Local Location Server and Directory	Required	5.3.2.7.2.5	Met
	LSC Transport Interface Functions	Required	5.3.2.7.2.7	Met
	LSC to PEI, AEI, and Operator Console Status Verification	Required	5.3.2.7.2.10	Partially Met (See notes 5 and 9.)
Line-Side Custom Features Interference	Conditional	5.3.2.7.2.11	Met	
	Loop Avoidance	Required	5.3.2.7.3	Met
7	<b>Call Connection Agent Requirements</b>			
	CCA-IWF Component	Required	5.3.2.9.2.1	Partially Met (See notes 10 and 11.)
	CCA MGC Component	Required	5.3.2.9.2.2	Met
	SG Component	Conditional	5.3.2.9.2.3	Not Tested (See note 10.)
	CCA-IWF Support for AS-SIP	Required	5.3.2.9.5.1	Met
	CCA-IWF Support for SS7	Conditional	5.3.2.9.5.2	Not Tested (See note 10.)
	CCA-IWF Support for PRI via MG	Required	5.3.2.9.5.3	Partially Met (See note 11.)
	CCA-IWF Support for CAS Trunks via MG	Conditional	5.3.2.9.5.4	Not Tested (See note 10.)
	CCA-IWF Support for PEI and AEI Signaling Protocols	Required	5.3.2.9.5.5	Partially Met (See note 12.)

**Table 2. SUT Capability Requirements and Functional Requirements Status (continued)**

CR/FR ID	Capability/ Function	Applicability (See note 1.)	UCR Reference	Status
7	<b>Call Connection Agent Requirements (continued)</b>			
	CCA-IWF Support for VoIP and TDM Protocol Interworking	Required	5.3.2.9.5.6	Met (See note 10.)
	CCA Preservation of Call Ringing State during Failure Conditions	Required	5.3.2.9.6	Met
	CCA Interactions with Transport Interface Functions	Required	5.3.2.10.3	Met
	CCA Interactions with the EBC	Required	5.3.2.10.4	Met
	CCA Support for Admission Control	Required	5.3.2.10.5	Met
	CCA Support for UFS	Required	5.3.2.10.6	Met
	CCA Support for IA	Required	5.3.2.10.7	Met (See note 13.)
	CCA Interaction with EIs	Required	5.3.2.10.10	Partially Met (See note 5.)
	CCA Support for AS Voice and Video	Required	5.3.2.10.11	Partially Met (See note 8.)
	CCA Interactions with Service control Functions	Required	5.3.2.10.12	Met
CCA Interworking between AS-SIP and SS7	Conditional	5.3.2.11	Not Tested (See note 10.)	
8	<b>MG Requirements</b>			
	Role of MG In LSC	Required	5.3.2.12.3.1	Met
	MG Support for ASAC	Required	5.3.2.12.4.1	Met
	MG and IA Functions	Required	5.3.2.12.4.2	Met (See note 13.)
	MG Interaction with Service Control Function	Required	5.3.2.12.4.3	Met
	MG Interactions with IP Transport Interface Functions	Required	5.3.2.12.4.4	Met
	MG-EBC interactions	Required	5.3.2.12.4.5	Met
	MG IP-Based PSTN Interface Requirements	Conditional	5.3.2.12.4.7	Not Tested
	MG Interaction with VoIP EIs	Required	5.3.2.12.4.8	Met (See note 5.)
	MG support for User Features and Services	Required	5.3.2.12.4.9	Met
	MG Interface to TDM	Required	5.3.2.12.5	Partially Met (See notes 10, 11, and 14.)
	MG Interface to TDM Allied and Coalition	Conditional	5.3.2.12.6	Not Tested
	MG Interface to TDM PSTN in US	Required	5.3.2.12.7	Partially Met (See notes 11 and 14.)
	MG Interfaces to TDM PSTN OCONUS	Required	5.3.2.12.8	Partially Met (See note 14.)
	MG Support for CCS7	Conditional	5.3.2.12.9	Not Tested (See note 10.)
	MG Support for ISDN PRI Trunks	Required	5.3.2.12.10	Partially Met (See note 11.)
	MG Support for CAS Trunks	Required	5.3.2.12.11	Met
	MG requirements for VoIP Internal Interfaces	Required	5.3.2.12.12	Met
	MG Echo Cancellation	Required	5.3.2.12.13	Met
MG Clock Timing	Required	5.3.2.12.14	Met	
MGC-MG CCA Functions	Required	5.3.2.12.15	Met	
MG ITU-T V.150.1	Required	5.3.2.12.16	Not Met (See note 15.)	
MG Preservation of Call Ringing during Failure	Required	5.3.2.12.17	Met	
9	<b>SG Requirements</b>			
	SG and CCS7 Network Interactions	Conditional	5.3.2.13.5.1	Not Tested
	SG Interactions with CCA	Conditional	5.3.2.13.5.2	Not Tested
	SG Interworking Functions	Conditional	5.3.2.13.5.3	Not Tested
10	<b>WWNDP Requirements</b>			
	WWNDP	Required	5.3.2.16	Met
	DSN WWNDP	Required	5.3.2.16.1	Met

**Table 2. SUT Capability Requirements and Functional Requirements Status (continued)**

CR/FR ID	Capability/ Function	Applicability (See note 1.)	UCR Reference	Status
11	<b>Commercial Cost Avoidance</b>			
	Commercial Cost Avoidance	Required	5.3.2.23	Not Tested (See note 15.)
12	<b>AS-SIP Based for External Devices (Voicemail, Unified Messaging, and Automated Receiving Devices)</b>			
	AS-SIP Requirements for External Interfaces	Conditional	5.3.2.24	Not Tested
13	<b>Precedence Call Diversion</b>			
	Precedence call Diversion	Required	5.3.2.25	Met
14	<b>Attendant Station Features</b>			
	Precedence and Preemption	Required	5.3.2.26.1	Not Tested (See note 9.)
	Call Display	Required	5.3.2.26.2	Not Tested (See note 9.)
	Class of Service Override	Required	5.3.2.26.3	Not Tested (See note 9.)
	Busy Override and Busy Verification	Required	5.3.2.26.4	Not Tested (See notes 9 and 17.)
	Night service	Required	5.3.2.26.5	Not Tested (See note 9.)
	Automatic Recall of Attendant	Required	5.3.2.26.6	Not Tested (See note 9.)
	Calls in Queue to the Attendant	Required	5.3.2.26.7	Not Tested (See notes 9 and 18.)
15	<b>AS-SIP Requirements</b>			
	SIP Requirements for AS-SIP Signaling Appliances and AS-SIP EIs	Required	5.3.4.7	Not Tested (See note 5.)
	SIP Session Keep-Alive Timer	Required	5.3.4.8	Met
	Session Description Protocol	Required	5.3.4.9	Met
	Precedence and Preemption	Required	5.3.4.10	Met
	Video Telephony – General Rules	Required	5.3.4.12	Not Met (See note 8.)
	Calling Services	Required	5.3.4.13	Met
	SIP Translation Requirements for Inter-working AS-SIP Signaling Appliances	Required	5.3.4.14	Met (See note 19.)
	Relevant Timers for the Terminating Gateway and the Originating Gateway	Required	5.3.4.15	Not Tested (See note 20.)
	SIP Requirements for Interworking AS-SIP Signaling Appliances	Required	5.3.4.16	Met
	Keep-Alive Timer Requirements for Interworking AS-SIP Signaling Appliances	Required	5.3.4.17	Met
Precedence and Preemption Extensions for Interworking AS-SIP Signaling Appliances	Required	5.3.4.18	Met	
	Supplementary Services	Required	5.3.4.19	Met
16	<b>IPv6 Requirements</b>			
	Product Requirements	Required	5.3.5.4	Partially Met (See notes 8 and 21.)
17	<b>NM</b>			
	LSC Management Function	Required	5.3.2.7.2.6	Met
	VVoIP NMS Interface Requirements	Required	5.3.2.4.4	Met
	General Management requirements	Required	5.3.2.17.2	Met
	Requirement for FCAPS Management	Required	5.3.2.17.3	Met
	NM requirements of Appliance Functions	Required	5.3.2.18	Met
	Accounting Management	Required	5.3.2.19	Partially Met (See note 22.)

**Table 2. SUT Capability Requirements and Functional Requirements Status (continued)**

**NOTES:**

1. The annotation of 'required' refers to a high-level requirement category. The applicability of each sub-requirement is provided in Reference (e), Enclosure 3.
2. The SUT does not support a "ping ring" notification. DISA adjudicated this as minor and changed this to optional in UCR 2013.
3. DTR 2 was requested to include the VG350 with Internetwork Operating System (IOS) 15.2(4)M1 as a certified SUT voice gateway in addition to the VG224 with IOS 15.1(4)M2. JITC conducted verification and validation (V&V) testing from 17 through 19 December 2012. JITC testing verified that the VG350 voice gateway with IOS 15.2(4)M1 worked properly as a voice gateway; however, it introduced one new discrepancy. The VG350 gateway crashes when Precedence Call Waiting is invoked subsequently on an analog line. The only affects a precedence call above ROUTINE and is only caused when a second call above ROUTINE is placed to the same line in succession. This discrepancy was fixed and successfully tested with DTR 7, which included 15.2(4)M3 on the VG350.
4. The SUT did not initially support the LSC failover requirements in UCR 2008, Change 2, paragraph 5.3.2.3.2, but demonstrated compliance during DTR 5 testing. DISA NS2 supported a multi-vendor failover test event at JITC from 8 through 19 October 2012 to newly drafted failover requirements in the UCR. The SUT met these new requirements, which DISA has stated will be included in the next version of the UCR (UCR 2013).
5. During the original interoperability test effort, the SUT only supported voice PEIs. The vendor did not support AEIs (voice or video). DISA accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact. During testing for DTR 1, the SUT successfully demonstrated the ability to support the video EIs shown in Table 1. The SUT proprietary video EIs are certified specifically with other certified CUCM Version 8.6.1 LSCs with proprietary video EIs intra-enclave using either IPv4 or IPv6 and inter-enclave using IPv4 only.
6. The SUT gateways equipped with the PVD3 modules do not support the ITU-T G.723 codec. DISA adjudicated this as minor and changed this to optional in UCR 2013.
7. The SUT softphone with Microsoft Windows Vista and Windows 7 OSs does not allow DSCP tagging per precedence level in accordance with UCR 2008, Change 2, Section 5.3.3.3.2. Microsoft Windows XP is the only OS that supports the five precedence levels. DISA adjudicated this as minor since all voice is queued together in the four-queue model currently used in deployed ASLANs.
8. During the original test, the SUT did not offer a video PEI. DTR 1 was requested to include the Cisco proprietary video EIs. The SUT with the Cisco proprietary video EIs met all requirements with the following minor exceptions:
  - a. The Cisco proprietary video EI does not properly tag IPv6 traffic for call signaling. All IPv6 media packets are tagged properly. Call Signaling packets are tagged at 0 (best effort), with the exception of Registration Admission Status (RAS) signaling packets. In addition the SUT is not able to set the DSCP tag any value 0-63 for these packets as required by the reference.
  - b. The Cisco proprietary video EI is not able to disable all IPv6 services in the IPv6 stack, only IPv6 media and signaling. In accordance with Reference (c), all nodes and interfaces that are IPv6-capable must be carefully configured and verified that the IPv6 stack is disabled until it is deliberately enabled as part of a deliberate transition strategy
  - c. The Cisco proprietary video EI does not properly tag the DSCP for video and voice media. This applies to both IPv4 and IPv6 signaling and media IP traffic. The DSCP value can be set; however, all traffic is tagged with the same DSCP value.  
DISA accepted the vendor's PoA&M and adjudicated these three discrepancies as minor.
  - d. The SUT VCS partially complies with support of IPv6 scoped address architecture. The VCS does not support link-local address. The Office of the Department of Defense (DoD) Chief Information Officer (CIO) provided a waiver of these two IPv6 requirements that have minimal or no impact on interoperability.
9. The SUT Operator Console/Attendant Station was not tested; however the vendor submitted an LoC for the requirements.
10. The SUT met T1 ISDN PRI (ANSI T1.619a and ANSI T1.607), E1 PRI (ITU-T Q.931), and T1 CAS DTMF IWF requirements, which is all of the certified TDM interfaces. The SUT met the E1 PRI (ITU-T Q.955.3) interface requirements during testing for DTR 3.
11. The SUT does not support NFAS on the T1 ISDN PRI interface. Although this is conditional for DSN connectivity, it is required for PSTN connectivity. DISA adjudicated this as minor and stated the intent to change this to conditional in the next version of the UCR (UCR 2013).
12. The SUT met PEI CCA-IWF requirements. The SUT does not support AEIs. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
13. The security requirements are tested by a DISA-led IA test team and published in a separate report, Reference (f).
14. The SUT must meet T1 PRI (T1.619a and NI2) IWF. The T1 CAS and T1 CCS7 IWF requirements are conditional. The SUT met T1 ISDN PRI (ANSI T1.619a and ANSI T1.607), E1 PRI (ITU-T Q.931), and T1 CAS DTMF IWF requirements. The SUT met the E1 PRI (ITU-T Q.955.3) interface requirements during testing for DTR 3.
15. The SUT does not properly handle ITU-T V.150 calls with the Avaya Communication Manager 6.0 and both vendors are working on the problem. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
16. The SUT does not support Commercial Cost Avoidance. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
17. The SUT does not fully comply with Busy Override and Busy Line Verification requirements. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
18. The SUT does not fully comply with attendant console queuing requirements. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
19. The SUT met this requirement with ANSI T1.619a ISDN PRI NI2 DSN and ISDN PRI NI2 PSTN TDM interfaces interworking with AS-SIP. This requirement was met with both testing and the vendor's LoC. The SUT does not support CCS7 TDM interface which is conditional for an LSC.

**Table 2. SUT Capability Requirements and Functional Requirements Status (continued)**

<b>NOTES (continued):</b>			
20. This requirement applies to gateways between AS-SIP and CCS7 links. Because CCS7 is a conditional requirement for LSCs and not supported by the SUT, this requirement was not tested.			
21. The vendor submitted an IPv6 LoC with noted discrepancies. The SUT does not support RFCs 4861 and 4862. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.			
22. The vendor submitted an NM LoC with noted discrepancies. The SUT does not comply with the requirement for the equipment impairment factor to be in accordance with ITU-T G.107. DISA adjudicated this as minor with a vendor POA&M to provide a MOS score. DISA changed this requirement to optional in UCR 2013. The SUT does not have the ability to transfer records to a removable physical storage media. DISA adjudicated this as minor and removed this requirement from UCR 2013.			
<b>LEGEND:</b>			
AEI	AS-SIP End Instrument	MGC	Media Gateway Controller
ANSI	American National Standards Institute	MLPP	Multi-Level Precedence and Preemption
APL	Approved Products List	MOS	Mean Opinion Score
AS	Assured Services	NFAS	Non Facility Associated Signaling
ASAC	Assured Services Admission Control	NI2	National ISDN Standard 2
ASLAN	Assured Services Local Area Network	NM	Network Management
AS-SIP	Assured Services Session Initiation Protocol	NMS	Network Management System
BRI	Basic Rate Interface	OCONUS	Outside the Continental United States
CAS	Channel Associated Signaling	OS	Operating System
CCA	Call Connection Agent	PBAS	Precedence Based Assured Services
CR	Capability Requirement	PEI	Proprietary End Instrument
CCS7	Common Channel Signaling	POA&M	Plan of Action and Milestones
DISA	Defense Information Systems Agency	PRI	Primary Rate Interface
DSCP	Differentiated Services Code Point	PSTN	Public Switched Telephone Network
DSN	Defense Switched Network	PVDM3	Packet Voice Digital Signal Processor Module 3
DSS1	Digital Subscriber Signaling 1	Q.931	Signaling Standard for ISDN
DTMF	Dual Tone Multi-Frequency	RFCs	Request for Comments
DTR	Desktop Review	SG	Signaling Gateway
E1	European Basic Multiplex Rate (2.048 Mbps)	SIP	Session Initiation Protocol
EBC	Edge Boundary Controller	SS	Softswitch
EI	End Instrument	SS7	Signaling System 7
FCAPS	Fault, Configuration, Accounting, Performance and Security	SUT	System Under Test
FR	Functional Requirement	T1	Digital Transmission Link Level 1 (1.544 Mbps)
IA	Information Assurance	T1.607	ISDN – Layer 3 Signaling Specification for Circuit Switched Bearer Service for DSS1
ID	Identification	T1.619a	SS7 and ISDN MLPP Signaling Standard for T1
ISDN	Integrated Services Digital Network	TDM	Time Division Multiplexing
IP	Internet Protocol	UC	Unified Capabilities
IPv6	Internet Protocol version 6	UCR	Unified Capabilities Requirements
ITU-T	International Telecommunication Union - Telecommunication Standardization Sector	UFS	User Features and Services
IWF	Interworking Function	US	United States
LoC	Letter of Compliance	V.150	Modem over Internet Protocol Networks
LSC	Local Session Controller	VoIP	Voice over Internet Protocol
Mbps	Megabits per second	VVoIP	Voice and Video over Internet Protocol
MG	Media Gateway	WAN	Wide Area Network
		WWNDP	Worldwide Numbering and Dialing Plan

5. No detailed test report was developed in accordance with the Program Manager's request. JITC distributes interoperability information via the JITC Electronic Report Distribution (ERD) system, which uses Unclassified-But-Sensitive Internet Protocol Router Network (NIPRNet) e-mail. More comprehensive interoperability status information is available via the JITC System Tracking Program (STP). The STP is accessible by .mil/gov users on the NIPRNet at <https://stp.fhu.disa.mil>. Test reports, lessons learned, and related testing documents and references are on the JITC Joint Interoperability Tool (JIT) at <http://jit.fhu.disa.mil> (NIPRNet). Information related to DSN testing is on the Telecom Switched Services Interoperability (TSSI) website at <http://jitc.fhu.disa.mil/tssi>. Due to the sensitivity of the information, the Information

JITC Memo, JTE, Extension of the Special Interoperability Test Certification of the Cisco Unified Communications Manager (CUCM) Version 8.6.1

Assurance Accreditation Package (IAAP) that contains the approved configuration and deployment guide must be requested directly through government civilian or uniformed military personnel from the Unified Capabilities Certification Office (UCCO), e-mail: [disa.meade.ns.list.unified-capabilities-certification-office@mail.mil](mailto:disa.meade.ns.list.unified-capabilities-certification-office@mail.mil).

6. The JITC point of contact is Capt Jonathan Kim, DSN 879-5182, commercial (520) 538-5182, FAX DSN 879-4347, or e-mail to [jonathan.s.kim.mil@mail.mil](mailto:jonathan.s.kim.mil@mail.mil). JITC's mailing address is P.O. Box 12798, Fort Huachuca, AZ 85670-2798. The UCCO tracking number for the SUT is 1108301.

FOR THE COMMANDER:



Enclosure a/s

for RICHARD A. MEADOR  
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U.S. Strategic Command, J665

US Navy, OPNAV N2/N6FP12

US Army, DA-OSA, CIO/G-6 ASA(ALT), SAIS-IOQ

US Air Force, A3CNN/A6CNN

US Marine Corps, MARCORSSYSCOM, SIAT, A&CE Division

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NSG Interoperability Assessment Team

DOT&E, Netcentric Systems and Naval Warfare

Medical Health Systems, JMIS IV&V

HQUSAISEC, AMSEL-IE-IS

UCCO

## **ADDITIONAL REFERENCES**

- (c) Office of the Assistant Secretary of Defense, "Department of Defense Unified Capabilities Requirements 2008, Change 2," December 2010
- (d) Joint Interoperability Test Command, "Unified Capabilities Test Plan (UCTP)," Draft
- (e) Joint Interoperability Test Command, Memo, JTE, "Special Interoperability Test Certification of the Cisco Unified Communications Manager (CUCM) Version 8.6.1," 6 September 2012
- (f) Joint Interoperability Test Command, "Information Assurance (IA) Assessment of Cisco Unified Communications Manager (CUCM), Version 8.6.1, (TN 1108301)," Draft