



DEFENSE INFORMATION SYSTEMS AGENCY

P. O. BOX 549
FORT MEADE, MARYLAND 20755-0549

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

10 Dec 12

MEMORANDUM FOR DISTRIBUTION

SUBJECT: Extension of the Special Interoperability Test Certification of the Avaya Aura[®] Application Server (AS) 5300 with Software Release 3.0 Local Session Controller (LSC)

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 (g), see Enclosure

1. References (a) and (b) establish Defense Information Systems Agency (DISA) Joint Interoperability Test Command (JITC), as the responsible organization for interoperability test certification. Reference (c) further establishes JITC as the interoperability Certification Authority (CA) for all Unified Capabilities (UC) products.

2. The Avaya Aura[®] AS 5300 with Software Release 3.0; hereinafter referred to as the System Under Test (SUT), is certified for joint use within the Defense Information System Network (DISN) as an LSC. The SUT is certified in the United States, including the Continental United States (CONUS), Alaska, Hawaii, and U.S. Caribbean and Pacific Territories. Although the SUT supports European Basic Multiplex Rate (E1) interfaces, they were not tested and are not covered under this certification. Therefore, the SUT is not certified for joint use outside CONUS in European Telecommunications Standards Institute (ETSI)-compliant countries. The fielding of the SUT is limited to IP version 4 (IPv4) across the DISN. Although the SUT supports Internet Protocol version 6 (IPv6), it was not tested inter-enclave because of a limitation of the test network Edge Boundary Controller (EBC) which currently does not support end-to-end IPv6. Therefore, IPv6 is not covered under this certification. JITC will verify inter-enclave IPv6 capabilities of the SUT prior to amending the certification to include the capability. Intra-enclave use of IPv4 and IPv6 is authorized for use. DISA adjudicated all open Test Discrepancy Reports (TDRs) to have a minor operational impact. 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's Plan of Action and Milestones (POA&M), which will address all new critical TDRs within 120 days of identification. Testing was conducted using LSC product requirements derived from the Unified Capabilities Requirements (UCR), Reference (d), and LSC test procedures, Reference (e). 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 three years from the date of the original UC Approved Products List (APL) memorandum (2 October 2012); the expiration date is 2 October 2015.

3. The extension of this certification is based upon Desktop Review (DTR) 1. The original certification, documented in Reference (f), is based on interoperability testing conducted by JITC, DISA adjudication of open TDRs, review of the vendor's Letters of Compliance (LoC), and DISA CA approval of the Information Assurance (IA) configuration. Interoperability testing was conducted by JITC, Fort Huachuca, Arizona, from 27 February through 20 April 2012. Additional interoperability testing was conducted from 14 May to 8 June 2012 to address test procedures not completed during the initial test window as well as new firmware on the SUT's end instruments. Review of the vendor's LoC was completed on 11 July 2012. DISA adjudication of outstanding TDRs was completed on 2 August 2012. Additional interoperability testing was conducted from 6 through 8 August 2012, which resulted in the successful demonstration of International Telecommunication Union - Telecommunication Standardization Sector T.38 fax functionality. The DISA CA has reviewed the IA Assessment Report for the SUT, Reference (f), and based on the findings in the report has provided a positive recommendation on 27 September 2012. This DTR was requested to add the IBM HS22 Blade Server hardware with the currently certified HP DL360 and IBM x3550 servers for use as the Avaya Media Server and Session Initiation Protocol (SIP) Core Operations, Administration, Maintenance, and Provisioning (OAM&P)/LSC Session Manager (SESM). The IBM HS22 Blade Server functions in the same capacity as the currently certified HP DL360 and IBM x3550 Avaya Media Server and SIP Core OAM&P/LSC SESM servers. This is a hardware addition only; there is no difference in software. JITC determined that testing was not required because past experience has shown that different vendor server products with similar processor speeds and memory size as the certified server products almost always perform equivalently and have no impact on the certified interoperability features and functions. Therefore, based on risk evaluation, JITC approves this DTR without conducting a verification & validation test. The IA posture has not changed. Therefore, the original IA approval applies to this DTR.

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

Table 1. SUT Interface Interoperability Status

Interface	Critical	UCR Reference	Threshold CR/FR Requirements ¹	Status	Remarks
Line Interfaces					
10Base-X	Yes	5.3.2.6.3	2, 4, 10, 13, 16, and 19	Certified	Met threshold CRs/FRs for IEEE 802.3i and 802.3j with the SUT PEIs.
100Base-X	Yes	5.3.2.6.3	2, 4, 10, 13, 16, and 19	Certified	Met threshold CRs/FRs for IEEE 802.3u with the SUT PEIs.
1000Base-X	No	5.3.2.6.3	2, 4, 10, 13, 16, and 19	Certified	Met threshold CRs/FRs for IEEE 802.3z with the SUT PEIs.
2-wire analog	Yes	5.3.2.6.1.6	2, 4, 10, 13, and 19	Certified	Met threshold CRs/FRs for 2-wire analog interfaces with the SUT IAD.
ISDN BRI	No	5.3.2.6.1.8	2, 4, 10, 13, and 19	Not Tested	This interface is not supported by the SUT and is not required for an LSC.
External Interfaces					
10Base-X	No ²	5.3.2.4.2	1, 2, 3, 6, 7, 8, 10, 11, 13, 15, 16, 17, and 19	Certified	Met threshold CRs/FRs for IEEE 802.3i and 802.3j for the AS-SIP trunk.
100Base-X	No ²	5.3.2.4.2	1, 2, 3, 6, 7, 8, 10, 11, 13, 15, 16, 17, and 19	Certified	Met threshold CRs/FRs for IEEE 802.3u for the AS-SIP trunk.

Table 1. SUT Interface Interoperability Status (continued)

Interface	Critical	UCR Reference	Threshold CR/FR Requirements ¹	Status	Remarks
External Interfaces (continued)					
1000Base-X	No ²	5.3.2.4.2	1, 2, 3, 6, 7, 8, 10, 11, 13, 15, 16, 17, and 19	Certified	Met threshold CRs/FRs for IEEE 802.3z for the AS-SIP trunk.
ISDN T1 PRI ANSI T1.619a	Yes	5.3.2.4.3	2, 3, 7, 8, 10, 13, and 19	Certified	Met threshold CRs/FRs. This interface provides legacy DSN and TELEPORT connectivity.
ISDN T1 PRI NI-2	Yes	5.3.2.4.3	2, 3, 7, 8, 10, 13, and 19	Certified	Met threshold CRs/FRs. This interface provides PSTN connectivity.
T1 CCS7 ANSI T1.619a	No	5.3.2.12.9	2, 3, 7, 8, 10, 13, and 19	Not Tested	Although this interface is offered by the SUT, it was not tested. This interface is not certified by JITC and is not required for an LSC.
T1 CAS	No	5.3.2.12.11	2, 3, 7, 8, 10, 13, and 19	Not Tested	This interface is not offered by the SUT and it is not required for an LSC.
E1 PRI ITU-T Q.955.3	No ³	5.3.2.12.10	2, 3, 7, 8, 10, 13, and 19	Not Tested	Although this interface is offered by the SUT, it was not tested. This interface is not certified by JITC and is not required for an LSC.
E1 PRI ITU-T Q.931	No ³	5.3.2.12.10	2, 3, 7, 8, 10, 13, and 19	Not Tested	Although this interface is offered by the SUT, it was not tested. This interface is not certified by JITC and is not required for an LSC.
NM Interfaces					
10Base-X	No ²	5.3.2.4.4 5.3.2.7.2.8	17, 18, and 19	Certified	Met threshold CRs/FRs. Verified via LoC.
100Base-X	No ²	5.3.2.4.4 5.3.2.7.2.8	17, 18, and 19	Certified	Met threshold CRs/FRs. Verified via LoC.
1000Base-X	No ²	5.3.2.4.4 5.3.2.7.2.8	17, 18, and 19	Certified	Met threshold CRs/FRs. Verified via LoC.
NOTES:					
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 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.					
LEGEND:					
10Base-X	10 Mbps Ethernet		ISDN	Integrated Services Digital Network	
100Base-X	100 Mbps Ethernet		ITU-T	International Telecommunication Union - Telecommunication Standardization Sector	
1000Base-X	1000 Mbps Ethernet		JITC	Joint Interoperability Test Command	
802.3i	10 Mbps twisted pair media for 10Base-X networks		LoC	Letter of Compliance	
802.3j	10 Mbps fiber media for 10Base-X networks		LSC	Local Session Controller	
802.3u	100BASE-TX, 100BASE-T4, 100BASE-FX Fast Ethernet at 100 Mbps with auto negotiation		Mbps	Megabits per second	
802.3z	Gigabit Ethernet Standard		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 Number 7		PSTN	Public Switched Telephone Network	
CR	Capability Requirement		Q.931	Signaling Standard for ISDN	
DSN	Defense Switched Network		Q.955.3	ISDN Signaling Standard for E1 MLPP	
E1	European Basic Multiplex Rate (2.048 Mbps)		SS7	Signaling System 7	
FR	Functional Requirement		SUT	System Under Test	
Gbps	Gigabits per second		T1	Digital Transmission Link Level 1 (1.544 Mbps)	
IAD	Integrated Access Device		T1.619a	SS7 and ISDN MLPP Signaling Standard for T1	
ID	Identification		UCR	Unified Capabilities Requirements	
IEEE	Institute of Electrical and Electronics Engineers				

Table 2. SUT Capability Requirements and Functional Requirements Status

CR/FR ID	Capability/Function	Applicability ¹	UCR Reference	Status
1	Assured Services Product Features and Capabilities			
	DSCP Packet Marking	Required	5.3.2.2.1.4	Partially Met ²
	Voice Features and Capabilities	Required	5.3.2.2.2.1	Partially Met ^{3,4}
	Public Safety Features	Required	5.3.2.2.2.2	Partially Met ⁵
	ASAC – Open Loop	Required	5.3.2.2.2.3	Met
	Signaling Protocols	Required	5.3.2.2.3	Partially Met ⁶
	Signaling Performance	Conditional	5.3.2.2.4	Met
2	Registration, Authentication, and Failover			
	Registration	Required	5.3.2.3.1	Met
	Failover	Required	5.3.2.3.2	Not Met ^{7,8}
3	Product Physical, Quality, and Environmental Factors			
	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	Voice End Instruments			
	Tones and Announcements	Required	5.3.2.6.1.1	Partially Met ⁹
	Audio Codecs	Required	5.3.2.6.1.2	Partially Met ¹⁰
	VoIP PEI or AEI Audio Performance Requirements	Required	5.3.2.6.1.3	Partially Met ^{6,11}
	VoIP Sampling Standard	Required	5.3.2.6.1.4	Met
	Authentication to LSC	Required	5.3.2.6.1.5	Met
	Analog Telephone Support	Required	5.3.2.6.1.6	Partially Met ^{12,13}
	Softphones	Conditional	5.3.2.6.1.7	Partially Met ²
	ISDN BRI	Conditional	5.3.2.6.1.8	Not Tested ¹⁴
5	Video End Instruments			
	Video End Instrument	Required	5.3.2.6.2	Partially Met ^{15,2}
	Display Messages, Tones, and Announcements	Required	5.3.2.6.2.1	Partially Met ^{15,2}
	Video Codecs (Including Associated Audio Codecs)	Required	5.3.2.6.2.2	Partially Met ^{15,2}
6	LSC Requirements			
	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 IP PEI, AEI, and Operator Console Status Verification	Required	5.3.2.7.2.10	Not Met ^{6,16}
	Line-Side Custom Features Interference	Conditional	5.3.2.7.2.11	Met
	Loop Avoidance	Required	5.3.2.7.3	Not Met ¹⁷
7	Call Connection Agent Requirements			
	CCA IWF Component	Required	5.3.2.9.2.1	Met
	CCA MGC Component	Required	5.3.2.9.2.2	Met
	SG Component	Conditional	5.3.2.9.2.3	Not Tested ¹⁴
	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 ¹⁴
	CCA-IWF Support for PRI via MG	Required	5.3.2.9.5.3	Met
	CCA-IWF Support for CAS Trunks via MG	Conditional	5.3.2.9.5.4	Not Tested ¹⁴
	CCA-IWF Support for PEI and AEI Signaling Protocols	Required	5.3.2.9.5.5	Partially Met ^{6,15,16}
	CCA-IWF Support for VoIP and TDM Protocol Interworking	Required	5.3.2.9.5.6	Met
	CCA Preservation of Call Ringing State during Failure Conditions	Required	5.3.2.9.6	Not Met ¹⁸
	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

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

CR/FR ID	Capability/Function	Applicability ¹	UCR Reference	Status
7	Call Connection Agent Requirements (continued)			
	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
	CCA Interaction with VoIP EIs	Required	5.3.2.10.10	Partially Met ^{6, 15, 16}
	CCA Support for AS Voice and Video	Required	5.3.2.10.11	Met ^{6, 11, 15}
	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 ¹⁴
8	MG Requirements			
	Role of MG In LSC	Required	5.3.2.12.3.1	Partially Met ¹⁹
	MG Support for ASAC	Required	5.3.2.12.4.1	Met
	MG and IA Functions	Required	5.3.2.12.4.2	Met
	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 ¹⁴
	MG Interaction with VoIP EIs	Required	5.3.2.12.4.8	Partially Met ⁶
	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	Met
	MG Interface to TDM Allied and Coalition	Conditional	5.3.2.12.6	Not Tested ²⁰
	MG Interface to TDM PSTN in U.S.	Required	5.3.2.12.7	Met
	MG Interfaces to TDM PSTN OCONUS	Conditional	5.3.2.12.8	Not Tested ²⁰
	MG Support for CCS7	Conditional	5.3.2.12.9	Not Tested ¹⁴
	MG Support for ISDN PRI Trunks	Required	5.3.2.12.10	Met
	MG Support for CAS Trunks	Conditional	5.3.2.12.11	Not Tested ¹⁴
	MG requirements for VoIP Internal Interfaces	Required	5.3.2.12.12	Met ²¹
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 ^{12, 13}	
MG Preservation of Call Ringing during Failure	Required	5.3.2.12.17	Not Met ^{18, 22}	
9	SG Requirements			
	SG and CCS7 Network Interactions	Conditional	5.3.2.13.5.1	Not Tested ¹⁴
	SG Interactions with CCA	Conditional	5.3.2.13.5.2	Not Tested ¹⁴
	SG Interworking Functions	Conditional	5.3.2.13.5.3	Not Tested ¹⁴
10	WWNDP Requirements			
	WWNDP	Required	5.3.2.16	Met
	DSN WWNDP	Required	5.3.2.16.1	Partially Met ²³
11	Commercial Cost Avoidance			
	Commercial Cost Avoidance	Required	5.3.2.23	Not Tested ²⁴
12	AS-SIP Based for External Devices (Voicemail, Unified Messaging, and Automated Receiving Devices)			
	AS-SIP Requirements for External Interfaces	Conditional	5.3.2.24	Not Tested ¹⁴
13	Precedence Call Diversion			
	Precedence Call Diversion	Required	5.3.2.25	Partially Met ¹⁶

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

CR/FR ID	Capability/Function	Applicability ¹	UCR Reference	Status
14	Attendant Station Features			
	Precedence and Preemption	Required	5.3.2.26.1	Not Met ¹⁶
	Call Display	Required	5.3.2.26.2	Not Met ¹⁶
	Class of Service Override	Required	5.3.2.26.3	Not Met ¹⁶
	Busy Override and Busy Verification	Required	5.3.2.26.4	Not Met ¹⁶
	Night service	Required	5.3.2.26.5	Not Met ¹⁶
	Automatic Recall of Attendant	Required	5.3.2.26.6	Not Met ¹⁶
	Calls in Queue to the Attendant	Required	5.3.2.26.7	Not Met ¹⁶
15	RTS Routing Database Requirements			
	LSC to LRDB Interface: DB Queries for CCA	Required	5.3.2.28.3	Not Tested ²⁴
	CCA Query from LSC	Required	5.3.2.28.3.1	Not Tested ²⁴
	DB Response When Commercial Number is Not Found	Required	5.3.2.28.3.3	Not Tested ²⁴
	LSC to MRDB Interface: DB Updates for CCA and HR	Required	5.3.2.28.4	Not Tested ²⁴
	LDAP Update Operations	Required	5.3.2.28.4.1	Not Tested ²⁴
	RTS Routing DB "Opt Out" for LSC End Users	Required	5.3.2.28.4.2	Not Tested ²⁴
	Request Processing	Required	5.3.2.28.5.2.3	Not Tested ²⁴
	Client Time-Out	Required	5.3.2.28.5.2.3.1	Not Tested ²⁴
	Data Caching	Required	5.3.2.28.5.2.4.2	Not Tested ²⁴
	Failover Procedures	Required	5.3.2.28.5.2.5	Not Tested ²⁴
	MRDB Failover	Required	5.3.2.28.5.2.5.1	Not Tested ²⁴
	LRDB Failover	Required	5.3.2.28.5.2.5.2	Not Tested ²⁴
Alarms	Required	5.3.2.28.6.3	Not Tested ²⁴	
Logs	Required	5.3.2.28.6.4	Not Tested ²⁴	
	Performance Monitoring	Conditional	5.3.2.28.6.7	Not Tested ²⁴
16	AS-SIP Requirements			
	SIP Requirements for AS-SIP Signaling Appliances and AS-SIP EIs	Required	5.3.4.7	Partially Met ⁶
	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	Met
	Calling Services	Required	5.3.4.13	Met
	SIP Translation Requirements for Inter-working AS-SIP Signaling Appliances	Required	5.3.4.14	Met
	Relevant Timers for the Terminating Gateway and the Originating Gateway	Required	5.3.4.15	Not Tested ²⁵
	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
17	IPv6 Requirements			
	Product Requirements	Required	5.3.5.4	Not Met ²⁶
18	NM Requirements			
	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	Partially Met ^{27, 28}
	NM requirements of Appliance Functions	Required	5.3.2.18	Met
	Accounting Management	Required	5.3.2.19	Met
19	Information Assurance			
	Information Assurance Requirements	Required	5.4	Met ²⁹

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 Enclosure 3.
2. The SUT softphones do not support a distinct DSCP tag for each of the five precedence levels. Only one DSCP tag is supported for all precedence levels. This is a limitation of the operating system on the softphones (Microsoft Windows 7) and cannot be changed by the vendor. DISA adjudicated this as minor since all voice is queued together in the four-queue model currently used in deployed ASLANs.
3. The SUT does not support a reminder ring notification with Call Forwarding Variable. DISA adjudicated this as minor and stated the intent to change this requirement to conditional in the next version of the UCR.
4. When the SUT is in a call with the Cisco Unified Communications Manager 8.0(2), the SUT IP EIs do not release a call from hold and the call cannot be resumed. DISA adjudicated this as minor with the vendor POA&M to resolve this issue by 17 October 2012.
5. The SUT allows the preemption of a 911 caller and the 911 operator. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
6. Testing with the Teo AEI was unable to be completed due to issues with TLS and therefore the SUT is not certified with non-proprietary AEIs. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
7. When the SUT fails over from the primary SESM to the secondary SESM, the SUT IP EI's configured to use IPv6 take approximately 10 to 15 minutes to register to the secondary SESM. After this time, the IP EI's do successfully register to the secondary SESM and gain full functionality. Also, the SUT IP EI's intermittently dropped active calls during the failover. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
8. The SUT does not fully support LSC dual-homing failover requirements. DISA adjudicated this discrepancy and determined that the UCR failover requirements are immature and require a rewrite. Avaya, in coordination with DISA NS2, has agreed to participate in a multi-vendor interoperability test event to test failover mechanisms between LSCs and SSs in the timeframe determined by DISA NS2 in order to address this discrepancy. The outcome of this event will help determine the path forward. DISA NS2 has agreed to a Condition of Fielding that the initial UC APL certification will not provide for failover capability and this certification is predicated on participation in and successful outcome from NS2 scheduled multi-vendor test event.
9. The SUT IAD EI's (Audiocodes 112/124) do not provide PNT during preempt for reuse scenarios. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
10. The vendor submitted LoC states the SUT does not support the ITU-T G.722.1 voice codec. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
11. The SUT PEI's were tested and met audio performance requirements.
12. The SUT does not support the ITU-T V.150.1 protocol. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
13. The vendor cannot dynamically invoke ITU-T T.38 and ITU-T V.150.1 in accordance with UCR 2008, change 3, section 5.3.2.12.16. The vendor stated this is a limitation of the Audiocodes gateway and requires an update before this can be tested. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
14. This interface or capability is a conditional requirement for an LSC and was not tested.
15. The SUT demonstrated video requirements via Softphone only, not PEIs (Proprietary Hard Video Phones) nor AEI video phones. The vendor did not provide a PEI or AEI video capability. This was previously adjudicated for another vendor by DISA to have a low operational impact because of the limited deployment of PEIs with video.
16. The SUT does not support an attendant console. DISA adjudicated this as minor and stated the intent to change this requirement to conditional in the next version of the UCR. Furthermore, the SUT meets all MLPP diversion requirements with an alternate DN in lieu of an attendant console in accordance with UCR 2008, Change 3, section 5.3.2.2.2.1.2.5.
17. The SUT is not capable of preventing or detecting and stopping hair-pin routing loops over ANSI T1.619a and commercial PRI trunk groups between a legacy switch and an LSC. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
18. The SUT allows AS-SIP sessions in a ringing state to fail when an internal failure occurs within the CCA. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
19. The AudioCodes M800 and M3K gateways do not allow a mix of PSTN/DSN trunk gateway configurations. Based on the vendor's POA&M from Release 2.0, the vendor stated this discrepancy would be fixed in AudioCodes version 6.02.054 and would be implemented in the AS 5300 Release 3 by 7 June 2012. However, Release 3 includes AudioCodes version 6.02A.043.001 and not 6.02A.054. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
20. This requirement states that the appliance suppliers should support TDM trunk groups on their MG product that can interconnect with NEs in U.S. allied and coalition partner networks worldwide or foreign country PTT networks (OCONUS) worldwide. This requirement is for interconnection with a foreign country. The SUT is certified for use in the U.S., including CONUS, Alaska, Hawaii, and U.S. Caribbean and Pacific Territories. Although the SUT supports E1 interfaces, they were not tested and are not covered under this certification.
21. The SUT MGs do not support analog trunks. DISA stated the intent to change analog trunks to optional for an LSC MG in the next version of the UCR.
22. The SUT MGs allow AS-SIP sessions in ringing state to fail during internal failure in MG. DISA adjudicated this as minor and stated the intent to remove this requirement in the next version of the UCR.
23. The SUT does not support domain directory. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
24. The vendor has an LDAP server which is covered under a separate Interoperability Certification listed on the UC APL; however, this LDAP feature was not tested with the Release 3.0. DISA adjudicated this as minor because it was tested with Release 2.0. This feature will be tested with Release 3.0 once the LDAP is installed at JITC.

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

NOTES (continued):

25. 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.
26. Per the vendor submitted LoC, the SUT does not properly support the following IPv6 requirements. The SUT does not support all DHCPv6 client messages and options. The SUT does not log all reconfigure events. The SUT SIP Core/Avaya Media Server does not allow disabling of duplicate address detection. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
27. The SUT is not fully compliant with following NM call detail records format requirements. The SUT does not provide a voice quality record at the completion of each voice session. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact. The SUT has the ability to send the records over a secure connection. However, the SUT does not have the ability to transfer records to a removable physical storage media. DISA adjudicated this as minor and stated the intent to change this requirement to conditional in the next version of the UCR.
28. Although the SUT supports destination code controls, the SUT does not play the correct announcement to the calling party IAW the reference. DISA has accepted and approved the vendor's POA&M and adjudicated this discrepancy as having a minor operational impact.
29. The IA requirements are tested by an IA test team and the results published in a separate report, Reference (g).

LEGEND:

AEI	AS-SIP End Instrument	LRDB	Local Routing Database
ANSI	American National Standards Institute	LSC	Local Session Controller
APL	Approved Products List	Mbps	Megabits per second
AS	Assured Services	MG	Media Gateway
ASAC	Assured Services Admission Control	MGC	Media Gateway Controller
ASLAN	Assured Services Local Area Network	MLPP	Multi-Level Precedence and Preemption
AS-SIP	Assured Services Session Initiation Protocol	MRDB	Master Routing Database
BRI	Basic Rate Interface	NE	Network Element
CAS	Channel Associated Signaling	NM	Network Management
CCA	Call Connection Agent	NMS	Network Management System
CCS7	Common Channel Signaling Number 7	NS2	Network Services
CONUS	Continental United States	OCONUS	Outside the Continental United States
CR	Capability Requirement	PBAS	Precedence Based Assured Services
DB	database	PEI	Proprietary End Instrument
DHCPv6	Dynamic Host Control Protocol for IPv6	PNT	Precedence Notification Tone
DISA	Defense Information Systems Agency	POA&M	Plan of Action and Milestones
DN	Directory Number	PRI	Primary Rate Interface
DSCP	Differentiated Services Code Point	PSTN	Public Switched Telephone Network
DSN	Defense Switched Network	PTT	Push-to-Talk
E1	European Basic Multiplex Rate (2.048 Mbps)	RTS	Real Time Services
EBC	Edge Boundary Controller	SESM	Session Manager
EI	End Instrument	SG	Signaling Gateway
FCAPS	Fault, Configuration, Accounting, Performance and Security	SIP	Session Initiation Protocol
FR	Functional Requirement	SS	Softswitch
G.722.1	ITU-T audio codec standard	SS7	Signaling System 7
HR	Hybrid Routing	SUT	System Under Test
IA	Information Assurance	T1	Digital Transmission Link Level 1 (1.544 Mbps)
IAD	Integrated Access Device	T1.619a	SS7 and ISDN MLPP Signaling Standard for T1
IAW	in accordance with	T.38	Fax over IP
ID	Identification	TDM	Time Division Multiplexing
IP	Internet Protocol	TLS	Transport Layer Security
IPv6	Internet Protocol version 6	UC	Unified Capabilities
ISDN	Integrated Services Digital Network	UCR	Unified Capabilities Requirements
ITU-T	International Telecommunication Union - Telecommunication Standardization Sector	UFS	User Features and Services
IWF	Interworking Function	U.S.	United States
JITC	Joint Interoperability Test Command	V.150	Modem over Internet Protocol Networks
LDAP	Lightweight Directory Access Protocol	VoIP	Voice over Internet Protocol
LoC	Letter of Compliance	VVoIP	Voice and Video over Internet Protocol
		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)

JITC Memo, JTE, Extension of the Special Interoperability Test Certification of the Avaya Aura® Application Server (AS) 5300 Local Session Controller (LSC), Software Release 3.0

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 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. All associated data is available on the DISA UCCO website located at <http://www.disa.mil/ucco/>.

6. The JITC point of contact is Capt Stéphane Arsenault, DSN 879-5269, commercial (520) 538-5269, FAX DSN 879-4347, or e-mail to Stephane.P.Arsenault.fm@mail.mil. JITC's mailing address is P.O. Box 12798, Fort Huachuca, AZ 85670-2798. The UCCO tracking number for the SUT is 1129302.

FOR THE COMMANDER:



for BRADLEY A. CLARK
Acting Chief
Battlespace Communications Portfolio

Enclosure a/s

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DISA/TEMC

DIA, Office of the Acquisition Executive

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) Department of Defense Instruction 8100.04, "DoD Unified Capabilities (UC)," 9 December 2010
- (d) Office of the Assistant Secretary of Defense, "Department of Defense Unified Capabilities Requirements 2008, Change 3," September 2011
- (e) Joint Interoperability Test Command, "Unified Capabilities Test Plan (UCTP)," Draft
- (f) Joint Interoperability Test Command, Memo, JTE, "Special Interoperability Test Certification of the Avaya Aura[®] Application Server (AS) 5300 with Software Release 3.0 Local Session Controller (LSC)," 27 September 2012
- (g) Joint Interoperability Test Command, "Information Assurance (IA) Assessment of Avaya Application Server (AS) 5300 Release (Rel.) 3.0 Local Session Controller (LSC) (Tracking Number 1129302)," Draft