



DEFENSE INFORMATION SYSTEMS AGENCY

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

MEMORANDUM FOR DISTRIBUTION

21 Jun 11

SUBJECT: Extension of the Special Interoperability Test Certification of Interactive Intelligence[®], Inc. Customer Interaction Center[™] (CIC) Release 3.0 Software Update (SU) 9

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 the Defense Information Systems Agency (DISA), Joint Interoperability Test Command (JITC), as the responsible organization for interoperability test certification.

2. The Interactive Intelligence[®], Inc. CIC Release 3.0 SU 9 is hereinafter referred to as the system under test (SUT). The SUT meets all of its critical interoperability requirements and is certified for joint use within the Defense Information System Network (DISN) as a Private Branch Exchange (PBX) 2. The PBX 2 switches have no Military Unique Features (MUFs) and can only serve Department of Defense (DoD), non-DoD, non-governmental, and foreign government users having no missions or communications requirement to ever originate or receive Command and Control (C2) communications. Since PBX 2s do not support MUF Requirements detailed in Reference (c), connectivity to the DSN is not authorized until a waiver is granted by the Chairman of the Joint Chiefs of Staff (CJCS) for each site in accordance with Reference (d). No other configurations, features, or functions, except those cited within this report, are certified by the JITC. This certification expires upon changes that could affect interoperability, but no later than three years from the date the DISA Field Security Operations (FSO) provided a positive Certification and Accreditation (CA) Recommendation.

3. The extension of this certification is based upon Desktop Review (DTR) 3. The original certification is based on interoperability testing, review of the vendor's Letters of Compliance (LoC), and FSO CA Recommendation. Interoperability testing of the SUT was conducted at JITC's Global Information Grid Network Test Facility at Fort Huachuca, Arizona, from 22 through 30 July 2010 and documented in Reference (e). Review of vendor's LoC was completed on 30 March 2011. The FSO provided a positive CA Recommendation on 5 April 2011 based on the security testing completed by DISA-led IA test teams and published in a separate report, Reference (f). The SUT was tested with the Sun Netra T2000, which is now manufacturer discontinued. This DTR was requested to include the Sun Netra T5220 as an acceptable

platform for the Audiocodes Element Management System (EMS). The Sun Netra T5220 includes integrated features and additional ports and cores. The improved processor is capable of 64 simultaneous execution threads that is double the processing and throughput capacity of the T2000 with a decrease in energy consumption and operating costs. Based on previous experience, hardware changes of this nature present a very low risk and do not change the interoperability results. Therefore, JITC approves this DTR. The Information Assurance (IA) posture has not changed. The original IA approval applies to this DTR.

4. The interoperability test summary of the SUT is indicated in Table 1. The PBX 2 Capability Requirements (CRs) and Feature Requirements (FRs) are listed in Table 2. This interoperability test status is based on the SUT’s ability to meet:

- a. DSN services for Network and Applications specified in Reference (c).
- b. PBX 2 interface and signaling requirements for trunks/lines specified in Reference (d) verified through JITC testing and/or vendor submission of LoC.
- c. PBX 2 CRs/FRs specified in Reference (d) verified through JITC testing and/or vendor submission of LoC.
- d. The overall system interoperability performance derived from test procedures listed in Reference (g).

Table 1. SUT Interoperability Test Summary

DSN Trunk Interfaces			
Interface & Signaling	Critical	Status	Remarks
T1 CAS (DTMF)	No ¹	Not Tested	This interface is not supported by the SUT and is not required for a PBX 2.
E1 CAS (DTMF)	No (Europe only)	Not Tested	This interface is not supported by the SUT and is not required for a PBX 2.
T1 ISDN PRI NI 1/2 (ANSI T1.607)	No ¹	Certified	Met all critical CRs and FRs with the following minor exception: Secure voice calls fail at random intervals. ²
E1 ISDN PRI (ITU-T Q.931)	No (Europe only)	Not Tested	This interface is not supported by the SUT and is not required for a PBX 2.
DSN Line Interfaces			
Interface & Signaling	Critical	Status	Remarks
2-Wire Analog (GR-506-CORE)	Yes	Certified	Met all critical CRs and FRs.
ISDN BRI NI 1/2	No	Not Tested	This interface is not supported by the SUT and is not required for a PBX 2.
2-Wire Proprietary Digital	No	Not Tested	This interface is not supported by the SUT and is not required for a PBX 2.
VoIP (Ethernet IEEE 802.3u)	No	Certified	Met all critical CRs and FRs with the following minor exceptions: The SUT did not meet the IPv6 capability requirements. ³ The SUT incorrectly tagged OAM traffic. ⁴ The SUT does not support proper VLAN tagging with IP phones. The SUT is not certified for shared access. ⁵

Table 1. SUT Interoperability Test Summary (continued)

DSN Features and Capabilities				
Feature/Capability	Critical	Status	Remarks	
Common Features	Yes	Certified	Met all critical CRs and FRs.	
Attendant	No	Certified	Met all critical CRs and FRs.	
Public Safety	Yes	Certified	Met all critical CRs and FRs for the basic 911.	
Call Processing	Yes	Certified	Met all critical CRs and FRs.	
ISDN Services	No	Certified	Met all critical CRs and FRs with the T1 ISDN PRI interface.	
Synchronization	Yes	Certified	Met all critical CRs and FRs.	
VoIP System	No	Certified	Met all critical CRs and FRs with the following minor exception: The SUT did not meet the IPv6 capability requirements. ³	
Security	Yes	Certified	See note 6.	
Network Gateways				
Gateway	Interface & Signaling	Critical	Status	Remarks
PSTN	T1 CAS (DTMF)	No ¹	Not Tested	This interface is not supported by the SUT and is not required for a PBX 2.
	E1 CAS (DTMF)	No (Europe only)	Not Tested	This interface is not supported by the SUT and is not required for a PBX 2.
	T1 ISDN PRI NI 1/2 (ANSI T1.607)	No ¹	Certified	Met all critical CRs and FRs.
	E1 ISDN PRI (ITU-T Q.931)	No (Europe only)	Not Tested	This interface is not supported by the SUT and is not required for a PBX 2.
	Ground Start Line (GR-506-CORE)	No	Not Tested	This interface is not supported by the SUT and is not required for a PBX 2.
	Loop Start Line (GR-506-CORE)	No	Certified	Met all critical CRs and FRs with an LOC.
NOTES:				
1 The UCR 2008 does not specify a required interface for a PBX 2. A PBX 2 switch must support at least T1 CAS or T1 PRI.				
2 After a non-secure call is established, the users can go secure. Secure calls fail at random intervals from two to eight minutes with no known explanation. After a secure call fails, the users can reestablish the non-secure call without hanging up by pressing the non-secure button. Since secure call capability is a conditional requirement for a PBX 2, there is minor operational impact.				
3 The SUT does not support IPv6. The vendor was granted a waiver by ASD-NII on 27 February 2011 with the following stipulation: The SUT must demonstrate IPv6 capability by 31 August 2011.				
4 The phones incorrectly used a different value for signaling. OAM traffic was either tagged at zero or 34 decimal. This was previously adjudicated by DISA as having a minor operational impact.				
5 Phones which offered shared access did not put voice traffic in a separate VLAN as required. The SUT is not certified for shared access, which is not required.				
6 Security is tested by DISA-led Information Assurance test teams and published in a separate report, Reference (f).				

Table 1. SUT Interoperability Test Summary (continued)

LEGEND:			
802.3u	Standard for carrier sense multiple access with collision detection at 100 Mbps	ISDN	Integrated Services Digital Network
ANSI	American National Standards Institute	ITU-T	International Telecommunication Union - Telecommunication Standardization Sector
ASD/NII	Assistant Secretary of Defense for Networks and Information Integration	LSSGR	Local Access and Transport Area (LATA) Switching Systems Generic Requirements
BRI	Basic Rate Interface	Mbps	Megabits per second
CAS	Channel Associated Signaling	NI 1/2	National ISDN Standard 1 or 2
CRs	Capability Requirements	OAM	Operations Administration and Management
DISA	Defense Information Systems Agency	PBX 2	Private Branch Exchange 2
DSN	Defense Switched Network	PRI	Primary Rate Interface
DSS1	Digital Subscriber Signaling 1	PSTN	Public Switched Telephone Network
DTMF	Dual Tone Multi-Frequency	Q.931	Signaling Standard for ISDN
E1	European Basic Multiplex Rate (2.048 Mbps)	SUT	System Under Test
FRs	Feature Requirements	T1	Digital Transmission Link Level 1 (1.544 Mbps)
GR	Generic Requirement	T1.607	ISDN Layer 3 Signaling Specification for Circuit Switched Bearer Service for DSS1
GR-506-CORE	LSSGR: Signaling for Analog Interfaces	UCR	Unified Capabilities Requirements
IEEE	Institute for Electrical and Electronics Engineers	VoIP	Voice over Internet Protocol
IPv6	Internet Protocol version 6		

Table 2. PBX 2 Requirements

DSN Trunk Interfaces				
Interface	Critical	Requirements Required or Conditional		References
T1 CAS (MFR1, DTMF, DP)	No		<ul style="list-style-type: none"> • Direct Inward Dialing (C) • National ISDN 1/2 Primary Access (C) • ITU-T ISDN Primary Access (Europe only) (C) • Normal Wink Start Operations (C) • Glare Operation (C) • Abnormal Wink Start (C) • Glare Resolution (C) • Call for Service Timing (C) • Guard Timing (C) • Satellite Interface (C) • Disconnect Control (C) • Reselect and Retrial (C) • Off-Hook Supervision Transition (C) • Dial-Pulse Signals (C) • DTMF Signaling (C) • DSN ISDN User-to-Network Signaling (C) 	<ul style="list-style-type: none"> • UCR Section 5.2.1.3.2 • UCR Section 5.2.1.3.4.1 • UCR Section 5.2.1.3.4.2 • UCR Section 5.2.4.3.3.1.1 • UCR Section 5.2.4.3.3.1.2 • UCR Section 5.2.4.3.3.2 • UCR Section 5.2.4.3.3.2.2 • UCR Section 5.2.4.3.5 • UCR Section 5.2.4.3.6 • UCR Section 5.2.4.3.7 • UCR Section 5.2.4.3.8 • UCR Section 5.2.4.3.9 • UCR Section 5.2.4.3.10 • UCR Section 5.2.4.4.1 • UCR Section 5.2.4.4.2 • UCR Section 5.2.4.7.1
E1 CAS (MFR1, DTMF, DP)	No (Europe only)	Trunking	<ul style="list-style-type: none"> • Application (C) • Physical Layer (C) • Data Link Layer (C) • Data Link Connection (C) • Peer-to-Peer Procedures of Data-Link Layer (C) • Layer 3 DSN User-to-Network Signaling (C) • DSN User-to-Network Signaling for Circuit-Switched Bearer Services (C) • Sequence of Messages for DSN Circuit-Switched Calls (C) • Message Functional Definition and Content (C) • General Message Format and Information Elements Coding (C) 	<ul style="list-style-type: none"> • UCR Section 5.2.4.7.1.1 • UCR Section 5.2.4.7.1.2 • UCR Section 5.2.4.7.1.3 • UCR Section 5.2.4.7.1.3.1 • UCR Section 5.2.4.7.1.3.2 • UCR Section 5.2.4.7.1.4 • UCR Section 5.2.4.7.1.4.2 • UCR Section 5.2.4.7.1.4.3 • UCR Section 5.2.4.7.1.4.4 • UCR Section 5.2.4.7.1.4.5
T1 ISDN PRI NI 1/2 (ANSI T1.607)	No		<ul style="list-style-type: none"> • Supplementary Services (C) • Transmission (R) • PCM-24 Digital Trunk Interface (R) • Interface Characteristics (R) • Supervisory Channel Associated Signaling (C) • Clear Channel Capability (C) • Alarm and Restoral Requirements (C) • PCM-30 Digital Trunk Interface (Europe only) (C) • Interoperation of PCM-24 and PCM-30 (C) • Analog Trunk Interface (C) • Integrated Digital Loop Carrier (C) 	<ul style="list-style-type: none"> • UCR Section 5.2.4.7.1.4.6 • UCR Section 5.2.5 • UCR Section 5.2.6.1 • UCR Section 5.2.6.1.1 • UCR Section 5.2.6.1.2 • UCR Section 5.2.6.1.3 • UCR Section 5.2.6.1.4 • UCR Section 5.2.6.2 • UCR Section 5.2.6.3 • UCR Section 5.2.6.4 • UCR Section 5.2.6.5
E1 ISDN PRI (ITU-T Q.931)	No (Europe only)	Voice	<ul style="list-style-type: none"> • MOS (R) • Secure calls (C) 	<ul style="list-style-type: none"> • UCR 5.2.12.8.2.1 • CJCSI 6215.01C
		Facsimile	<ul style="list-style-type: none"> • Analog: ITU-T T.4 (R) 	<ul style="list-style-type: none"> • DISR
		Data	<ul style="list-style-type: none"> • Modem (VBD) (R) • 56 kbps switched data (C: PRI only) • 64 kbps switched data (C: PRI only) • NX56 synchronous BER (C: PRI only) • NX64 synchronous BER (C: PRI only) • Secure data (STE/STU-III) (C) 	<ul style="list-style-type: none"> • CJCSI 6215.01C • UCR Section 5.2.2.9.6 • UCR Section 5.2.2.9.6 • UCR Section 5.2.2.9.6 • UCR Section 5.2.2.9.6 • CJCSI 6215.01C
		VTC	<ul style="list-style-type: none"> • ITU-T H.320 (R: PRI only) 	<ul style="list-style-type: none"> • FTR 1080B-2002

Table 2. PBX 2 Requirements (continued)

DSN Line Interfaces				
Interface	Critical	Requirements Required or Conditional		References
2-Wire Analog	Yes	Access	<ul style="list-style-type: none"> Individual Line (R) PBX Line (C) National ISDN 1/2 Basic Access (C) 	<ul style="list-style-type: none"> UCR Section 5.2.1.1.1 UCR Section 5.2.1.3.1 UCR Section 5.2.1.3.3
ISDN BRI NI 1/2	No		<ul style="list-style-type: none"> Analog Line (C) Loop Start Line (R: 2-Wire Analog only) Reverse Battery (C) S/T Reference Point (ISDN BRI) (C) 	<ul style="list-style-type: none"> UCR Section 5.2.1.3.5 UCR Section 5.2.4.2.1 UCR Section 5.2.4.3.1 UCR Section 5.2.4.7.1.2.1
2-Wire Proprietary Digital	No	Voice	<ul style="list-style-type: none"> MOS (R) Secure Calls (C) 	<ul style="list-style-type: none"> CJCSI 6215.01C CJCSI 6215.01C
		Facsimile	<ul style="list-style-type: none"> Analog: ITU-T T.4 (R) 	<ul style="list-style-type: none"> DISR
		Data	<ul style="list-style-type: none"> Modem (VBD) (R) Secure data (STE/STU-III) (C) 	<ul style="list-style-type: none"> CJCSI 6215.01C CJCSI 6215.01C
		VTC	<ul style="list-style-type: none"> ITU-T H.320 (R: BRI only) 	<ul style="list-style-type: none"> FTR 1080B-2002
DSN Features & Capabilities				
Feature/Capability	Critical	Requirements Required or Conditional		References
Common Features	Yes	<ul style="list-style-type: none"> Individual Lines (R) Call waiting (C) Three-way calling (C) Add-on transfer, conference calling, and call hold (C) Call Transfer Individual – All calls (C) Call Transfer - Internal Only (C) Call Transfer – Individual – Incoming Only/Add-On Consultation Hold – Incoming Call (C) Call Transfer – Outside (C) Call Transfer – Add-On Restricted Station (C) Call Transfer – Attendant (C) Call Hold (C) Conference Calling – Six Way Station Controlled (C) Call forwarding Variable (C) Call Forward Busy Line (C) Call Forwarding – Don't Answer – All Calls (C) Selective Call Forwarding (C) Call pick-up (C) 		<ul style="list-style-type: none"> UCR Section 5.2.4.7.1.2.1 UCR Section 5.2.1.1.5.1 UCR Section 5.2.1.1.6 UCR Section 5.2.1.1.7 UCR Section 5.2.1.1.7.1 UCR Section 5.2.1.1.7.2 UCR Section 5.2.1.1.7.3 UCR Section 5.2.1.1.7.3 UCR Section 5.2.1.1.7.4 UCR Section 5.2.1.1.7.5 UCR Section 5.2.1.1.7.6 UCR Section 5.2.1.1.7.7 UCR Section 5.2.1.1.7.8 UCR Section 5.2.1.1.8.1 UCR Section 5.2.1.1.8.2 UCR Section 5.2.1.1.8.3 UCR Section 5.2.1.1.8.4 UCR Section 5.2.1.1.9.1
Attendant	No	<ul style="list-style-type: none"> Attendant Features (C) 		<ul style="list-style-type: none"> UCR Section 5.2.1.2
Public Safety	Yes	<ul style="list-style-type: none"> Emergency Service (911) Caller (R) Emergency Service (911) Public Safety Answering Service (C) Enhanced Emergency Service (E911) (C) 		<ul style="list-style-type: none"> UCR Section 5.2.1.4.1.1 UCR Section 5.2.1.4.1.2 UCR Section 5.2.1.4.1.3
Call Processing	Yes	<ul style="list-style-type: none"> Origination Treatment (R) Originating Busy (R) Termination Treatment (R) Busy or Idle Status (C) Release Treatment (R) Interruption Treatment (R) Connections (R) Class of Service (C) E&M Lead Signaling States (C) 4-Wire Analog User Access Lines (C) 2-Wire User Access Lines (C) Interswitch and Intraswitch Dialing (C) Calling Name Delivery (C) Calling Number Delivery (C) Screening (C) 		<ul style="list-style-type: none"> UCR Section 5.2.3.1.1 UCR Section 5.2.3.1.1.1 UCR Section 5.2.3.1.2 UCR Section 5.2.3.1.2.1 UCR Section 5.2.3.1.3 UCR Section 5.2.3.1.4 UCR Section 5.2.3.1.5 UCR Section 5.2.3.1.6 UCR Section 5.2.3.3.1 UCR Section 5.2.3.3.2 UCR Section 5.2.3.3.3 UCR Section 5.2.3.5.1.2 UCR Section 5.2.3.5.1.8.1 UCR Section 5.2.3.5.1.8.2 UCR Section 5.2.3.5.8

Table 2. PBX 2 Requirements (continued)

DSN Features & Capabilities (continued)			
Feature/ Capability	Critical	Requirements Required or Conditional	References
ISDN Services	No	<ul style="list-style-type: none"> • BRI Access, Call Control and Signaling (C) • Uniform Interface Configuration for BRIs (C) • BRI Features (C) • PRI Access, Call Control and Signaling (C) • PRI Features (C) • Packet Data Features and Capabilities (C) 	<ul style="list-style-type: none"> • UCR Section 5.2.9.2 Table 5.2.9-1 • UCR Section 5.2.9.2 Table 5.2.9-2 • UCR Section 5.2.9.2 Table 5.2.9-3 • UCR Section 5.2.9.2 Table 5.2.9-4 • UCR Section 5.2.9.2 Table 5.2.9-5 • UCR Section 5.2.9.2 Table 5.2.9-6
Synchronization	Yes	<ul style="list-style-type: none"> • Line timing mode (C) • Internal Stratum 4 (R) • Synchronization Performance Monitoring Criteria (C) • DS1 Traffic Interfaces (C) • DS0 Traffic Interconnects (C) 	<ul style="list-style-type: none"> • UCR Section 5.2.10.1.1.2 • UCR Section 5.2.10.1.2.2 • UCR Section 5.2.10.2 • UCR Section 5.2.10.3 • UCR Section 5.2.10.4
Security	Yes	<ul style="list-style-type: none"> • GR-815, STIGs, and DoDI 8510.bb (DIACAP) (R) 	<ul style="list-style-type: none"> • UCR Section 3
VoIP System	No	<p>VoIP function is conditional. If VoIP is provided, all of the following requirements must be met:</p> <ul style="list-style-type: none"> • Voice Quality with MOS of 4.0 or better (R) • ITU-T G.711 PCM CODEC (R) • MLPP (C) • Security (R) • Network management (C) • System timing (R) • Latency ≤ 60 milliseconds (R) • IPv6 capable (R) • Service Class Tagging (R) 	<ul style="list-style-type: none"> • UCR section 5.2.12.8.2.1 • UCR section 5.2.12.8.2.2 • UCR section 5.2.12.8.2.3 • UCR section 5.2.12.8.2.4 • UCR section 5.2.12.8.2.5 • UCR section 5.2.12.8.2.6 • UCR section 5.2.12.8.2.7 • UCR section 5.2.12.8.2.8 • UCR section 5.2.12.8.2.9
Network Gateways			
Gateway	Critical	Requirements Required or Conditional	References
PSTN	No	<p>Trunking</p> <ul style="list-style-type: none"> • Positive Identification Control (C) • On-Netting (C) • Off-Netting (C) • Loop Start Line (C) • Ground Start Line (C) • Immediate Start (C) • Delay Dial (C) 	<ul style="list-style-type: none"> • CJCSI 6215.01C • CJCSI 6215.01C • CJCSI 6215.01C • UCR Section 5.2.4.2.1 • UCR Section 5.2.4.2.2 • UCR Section 5.2.4.3.2 • UCR Section 5.2.4.3.4

Table 2. PBX 2 Requirements (continued)

LEGEND:					
ANSI	American National Standards Institute	FTR	Federal Telecommunications Recommendation	PCM-24	Pulse Code Modulation - 24 Channels
BER	Bit Error Ratio	FTR 1080B	Video Teleconferencing Services	PCM-30	Pulse Code Modulation - 30 Channels
BRI	Basic Rate Interface	GR	Generic Requirement	PRI	Primary Rate Interface
C	Conditional	GR-815	Generic Requirements For Network Element/Network System (NE/NS) Security	PSTN	Public Switched Telephone Network
CAS	Channel Associated Signaling	H.320	Standard for Narrowband VTC	Q.931	Signaling Standard for ISDN Required
CJCSI	Chairman of the Joint Chiefs of Staff Instruction	ISDN	Integrated Services Digital Network	R	ISDN BRI 4-wire interface
DIACAP	DoD Information Assurance Certification and Accreditation Process	IT	Information Technology	S/T	Secure Terminal Equipment
DISR	DoD IT Standards Registry	ITU-T	International Telecommunication Union - Telecommunication Standardization Sector	STIGs	Security Technical Implementation Guides
DoD	Department of Defense			STU-III	Secure Telephone Unit -3rd generation
DoDI	DoD Instruction	kbps	kilobits per second	T1	Digital Transmission Link Level 1 (1.544 Mbps)
DP	Dial Pulse	Mbps	Megabits per second	T1.607	ISDN Layer 3 Signaling Specification for Circuit Switched Bearer Service for DSS1
DS0	Digital Signal Level 0	MFR1	Multi-Frequency Recommendation 1	T.4	Standardization of Group 3 facsimile terminals for document transmission
DS1	Digital Signal Level 1 (1.544 Mbps) (2.048 Mbps European)	MLPP	Multi-Level Precedence and Preemption	UCR	Unified Capabilities Requirements
DSN	Defense Switched Network	MOS	Mean Opinion Score	VBD	Variable bit data
DSS1	Digital Subscriber Signaling 1	NI 1/2	National ISDN Standard 1 or 2	VoIP	Voice over Internet Protocol
DTMF	Dual Tone Multi-Frequency	NX56	Data format restricted to multiples of 56 kbps	VTC	Video Teleconferencing
E1	European Basic Multiplex Rate (2.048 Mbps)	NX64	Data format restricted to multiples of 64 kbps		
E911	Enhanced 911 Service	PBX	Private Branch Exchange		
E&M	Ear and Mouth	PBX 2	Private Branch Exchange 2		

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 <https://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: ucco@disa.mil.

6. The JITC point of contact is Ms. Anita Mananquil, DSN 879-5164, commercial (520) 538-5164, FAX DSN 879-4347, or e-mail to anita.mananquil@disa.mil. The JITC's mailing address is P.O. Box 12798, Fort Huachuca, AZ 85670-2798. The tracking number for the SUT is 0918002.

FOR THE COMMANDER:

Enclosure a/s


for BRADLEY A. CLARK
Chief
Battlespace Communications Portfolio

Distribution (electronic mail):

Joint Staff J-6

Joint Interoperability Test Command, Liaison, TE3/JT1

Office of Chief of Naval Operations, CNO N6F2

Headquarters U.S. Air Force, Office of Warfighting Integration & CIO, AF/XCIN (A6N)

Department of the Army, Office of the Secretary of the Army, DA-OSA CIO/G-6 ASA (ALT), SAIS-IOQ

U.S. Marine Corps MARCORSYSCOM, SIAT, MJI Division I

DOT&E, Net-Centric Systems and Naval Warfare

U.S. Coast Guard, CG-64

Defense Intelligence Agency

National Security Agency, DT

Defense Information Systems Agency, TEMC

Office of Assistant Secretary of Defense (NII)/DOD CIO

U.S. Joint Forces Command, Net-Centric Integration, Communication, and Capabilities
Division, J68

Defense Information Systems Agency, GS23

ADDITIONAL REFERENCES

- (c) Chairman of the Joint Chiefs of Staff Instruction (CJCSI) 6215.01C, "Policy for Department of Defense Voice Services with Real Time Services (RTS)," 9 November 2007
- (d) Office of the Assistant Secretary of Defense, "Department of Defense Unified Capabilities Requirements 2008," 22 January 2009
- (e) Joint Interoperability Test Command Memo, JTE, "Special Interoperability Test Certification of Interactive Intelligence[®], Inc. Customer Interaction Center[™] (CIC) Release 3.0 Software Update (SU) 9," 6 May 2011
- (f) Joint Interoperability Test Command, Memo, "Information Assurance (IA) Assessment of Interactive Intelligence Corporation (ININ) Customer Interaction Center (CIC) Release (Rel.) 3.0 (TN0918002)," 6 April 2011
- (g) Joint Interoperability Test Command, "Defense Switched Network Generic Switch Test Plan (GSTP), Change 2," 2 October 2006