

## Avaya Solution & Interoperability Test Lab

# Application Notes for Nexidia ESI-Capture with Avaya Aura<sup>TM</sup> Communication Manager and Avaya Aura<sup>TM</sup> Application Enablement Services - Issue 1.0

## **Abstract**

These Application Notes describe the procedures for configuring Nexidia ESI-Capture to monitor and record calls placed to and from stations on Avaya Aura TM Communication Manager.

The ESI-Capture is an application, built upon Nexidia's Scalable Media Processing (SMP) Framework that captures calls processed by an Avaya VoIP solution and records them along with any associated metadata. The ESI-Capture is composed of the SMP, the Avaya stream control/capture extension, and the recording sink extension. The ESI-Capture interfaces with Avaya Aura TM Communication Manager through Avaya Aura Application Enablement Services, using TSAPI to associate recordings with important CTI information, and DMCC to acquire media. The system uses the DMCC Streaming capability to record extension, and inbound or outbound calls. Voice is recorded at the server in way format.

Information in these Application Notes has been obtained through DevConnect compliance testing and additional technical discussions. Testing was conducted via the DevConnect Program at the Avaya Solution and Interoperability Test Lab.

## 1. Introduction

These Application Notes describe a compliance-tested configuration comprised of an Avaya Aura TM Communication Manager, an Avaya Aura Application Enablement Services server (AES), and the ESI-Capture. The ESI-Capture is a subset of the Nexidia Enterprise Speech Intelligence (ESI) system which utilizes Speech Analytics Technology to provide a scalable, accurate, affordable and fast solution to analyze all recorded audio.

The ESI-Capture monitors, records, stores, and plays back phone calls for verification. The ESI-Capture uses TSAPI with an Application Enablement Services server to monitor stations to obtain recording triggers and call information. The ESI-Capture also uses the Device, Media and Call Control (DMCC) service with the Application Enablement Services server to register DMCC softphones that the ESI-Capture uses as recording ports.

## 1.1. Interoperability Compliance Testing

The interoperability compliance test included features and serviceability. The focus of the compliance testing was primarily on verifying the interoperability between Nexidia ESI-Capture, SIP Enablement Services, and Communication Manager.

## 1.2. Support

Technical support for the Nexidia ESI-Capture solution can be obtained by contacting Nexidia:

North/South America and Asia/PAC

Phone: (866) 355-1241

Email: support@nexidia.com

• Europe, Middle East and Africa Email: <a href="mailto:EMEAsupport@nexidia.com">EMEAsupport@nexidia.com</a>

## 2. Reference Configuration

**Figure 1** provides the test configuration used for the compliance test. Note that actual configurations may vary. The solution described herein is also extensible to other Avaya Servers and Media Gateways. An Avaya S8300 Server with an Avaya G450 Media Gateway was included during the test to provide an IP trunk between two Communication Manager systems.

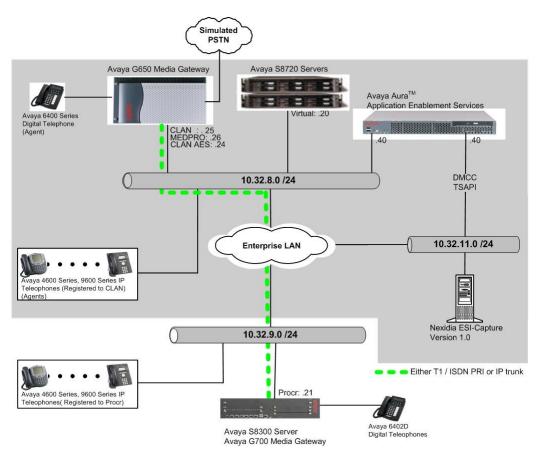


Figure 1: Sample Test Configuration for Nexidia ESI-Capture Solution

## 3. Equipment and Software Validated

The following equipment and software/firmware were used for the sample configuration provided:

Equipment	Software/Firmware
Avaya S8720 Server	Avaya Aura <sup>TM</sup> Communication Manager
	5.2.1 (R015x.02.1.016.4)
Avaya G650 Media Gateway	-
TN2312BP IP Server Interface	HW11 FW030
TN799DP C-LAN Interface	HW20 FW017
TN2302AP IP Media Processor	HW01 FW108
Avaya S8300 Server with Avaya G700 Media	Avaya Aura <sup>TM</sup> Communication Manager
Gateway	5.2.1 (R015x.02.1.016.4)
Avaya Aura <sup>TM</sup> Application Enablement Services	5.2 (r5-2-0-98-0)
Server	
Avaya 4600 Series IP Telephones	
4620SW (H.323)	2.9
4625SW (H.323)	2.9
Avaya 9600 Series IP Telephones	
9620 (H.323)	3.1
9630 (H.323)	3.1
9650 (H.323)	3.1
Avaya 6408D+ Digital Telephone	-
Nexidia ESI-Capture on Windows Server 2003	1.0
with Service Pack 2	

## 4. Configure Avaya Aura<sup>™</sup> Communication Manager

This section provides the procedures for configuring an ip-codec-set and ip-network region, switch connection and Computer Telephony Integration (CTI) links, monitored stations, and recording stations on Communication Manager. All the configuration changes in Communication Manager are performed through the System Access Terminal (SAT) interface. The highlights in the following screens indicate the values used during the compliance test.

## 4.1. Codec Configuration

Enter the **change ip-codec-set t** command, where **t** is a number between 1 and 7, inclusive.

```
change ip-codec-set 1
                                                               1 of
                                                                     2
                                                         Page
                       IP Codec Set
   Codec Set: 1
         Silence
   Audio
                          Frames
                                  Packet
             Suppression Per Pkt Size(ms)
   Codec
1: G.711MU
                   n
                                    20
2: G.729
```

## 4.2. IP Network Regions

During compliance testing, a C-LAN board dedicated for H.323 endpoint registration was assigned to IP network region 1. The Avaya IP Telephones and IP Softphones used by the ESI-Capture, registered with the C-LAN boards and were thus also assigned to IP network region 1. One consequence of assigning the aforementioned IP telephones, IP Softphones, and MedPro boards to a common IP network region is that the RTP traffic between them is governed by the same codec set. The second C-LAN board (CLAN-AES), which is dedicated for the AES server, was assigned to network region 2. The following screen shows only network region 1.

```
change ip-network-region 1
                                                                   Page
                                                                          1 of 19
                                IP NETWORK REGION
 Region: 1
                 Authoritative Domain:
Location:
   Name:
MEDIA PARAMETERS
                                 Intra-region IP-IP Direct Audio: yes
      Codec Set: 1
                                 Inter-region IP-IP Direct Audio: yes
  UDP Port Min: 2048
                                             IP Audio Hairpinning? n
  UDP Port Max: 3929
DIFFSERV/TOS PARAMETERS
                                           RTCP Reporting Enabled? y
Call Control PHB Value: 46 RTCP MONITOR SERVER PARAMETERS
Audio PHB Value: 46 Use Default Server Parameters'
                                 Use Default Server Parameters? y
       Video PHB Value: 46
802.1P/Q PARAMETERS
Call Control 802.1p Priority: 0
       Audio 802.1p Priority: 0
       Video 802.1p Priority: 5
                                       AUDIO RESOURCE RESERVATION PARAMETERS
H.323 IP ENDPOINTS
                                                           RSVP Enabled? n
 H.323 Link Bounce Recovery? y
Idle Traffic Interval (sec): 20
  Keep-Alive Interval (sec): 5
           Keep-Alive Count: 5
```

## 4.3. Configure Switch Connection and CTI Links between Avaya Aura<sup>™</sup> Communication Manager and Avaya Aura<sup>™</sup> Application Enablement Services

The AES server forwards CTI requests, responses, and events between the ESI-Capture and Communication Manager. The AES server communicates with Communication Manager over a switch connection link. Within the switch connection link, CTI links may be configured to provide CTI services to CTI applications such as the ESI-Capture. The following steps demonstrate the configuration of the Communication Manager side of the switch connection and CTI links. See **Section 5** for the details of configuring the AES side of the switch connection and CTI links.

Enter the **add cti-link m** command, where **m** is a number between 1 and 64, inclusive. Enter a valid extension under the provisioned dial plan in Communication Manager, set the Type field to **ADJ-IP**, and assign a descriptive Name to the CTI link.

add cti-link 4		Page	1 of	2
	CTI LINK			
CTI Link: 4				
Extension: 20006				
Type: ADJ-IP				
			COF	R: 1
Name: TSAPI				

Enter the **change node-names ip** command. In the compliance-tested configuration, the CLAN IP address was utilized for registering H.323 endpoints (Avaya IP Telephones, and IP Softphones, and AES Device, Media and Call Control API stations) and the CLAN-AES IP address was used for connectivity to Avaya AES.

change node-names	ip		Page	1 of	2
		IP NODE NAMES			
Name	IP Address				
CLAN	10.32.8.25				
CLAN-AES	10.32.8.24				
MEDPRO	10.32.8.26				
MEDPRO2	10.32.8.27				
S8300	10.32.10.21				
default	0.0.0.0				

Enter the **change ip-services** command. On **Page 1**, configure the Service Type field to **AESVCS** and the Enabled field to **y**. The Local Node field should be pointed to the **CLAN-AES** board that was configured previously in the IP NODE NAMES form in this section. During the compliance test, the default port was utilized for the Local Port field.

change ip-services Page						1 of	4
			IP SERVICES				
Service	Enabled	Local	Local	Remote	Remote		
Type		Node	Port	Node	Port		
AESVCS	У	CLAN-AES	8765				

On **Page 4**, enter the hostname of the AES server for the AE Services Server field. The server name may be obtained by logging in to the AES server using ssh, and running the command **uname –a**. Enter an alphanumeric password for the Password field. Set the Enabled field to **y**. The same password will be configured on the AES server in **Section 5.1**.



#### 4.4. Monitored Stations

Enter the **add station s** command, where **s** is an extension valid in the provisioned dial plan. During the compliance test, the following recorded stations were created.

- 22001 (Avaya 4625SW IP)
- 22002 (Avaya 9620 IP)
- 22003 (Avaya 9650 IP)
- 22004 (Avaya 9630 IP)
- 22007 (Avaya 6408D+)
- 22009 (Avaya IP Agent)

## 4.5. Recording Stations

Enter the **add station s** command, where **s** is an extension valid in the provisioned dial plan. On **Page 1** of the STATION form, set the Type field to an IP telephone set type, enter a descriptive Name, specify the Security Code, and make sure that the IP Softphone field is set to **y**. For the compliance test, recording stations from 23001 to 23023 were created.

```
change station 23001
                                                               Page 1 of 5
                                    STATION
Extension: 23001
                                        Lock Messages? n
                                                                      BCC: 0
                                        Security Code: *
                                                                       TN: 1
    Type: 4620
    Port: S00046
                                      Coverage Path 1:
                                                                       COR: 1
                                      Coverage Path 2:
                                                                       cos: 1
    Name: DMCC-1
                                      Hunt-to Station:
STATION OPTIONS
                                          Time of Day Lock Table:
             Loss Group: 19
                                  Personalized Ringing Pattern: 1
                                               Message Lamp Ext: 23001
       Speakerphone: 2-way
Display Language: english
                                           Mute Button Enabled? y
                                               Expansion Module? n
Survivable GK Node Name:
         Survivable COR: internal
                                              Media Complex Ext:
  Survivable Trunk Dest? y
                                                    IP SoftPhone? y
                                              IP Video Softphone? n
                                             Customizable Labels? y
```

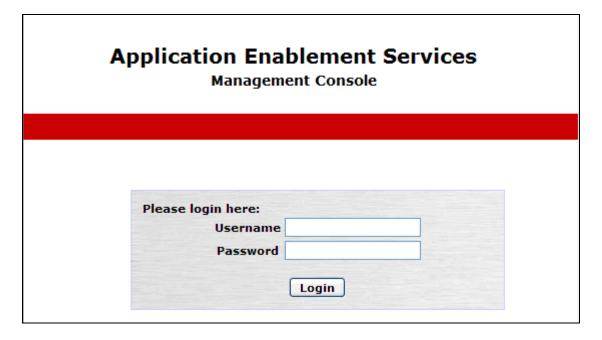
## 5. Configure Avaya Aura<sup>™</sup> Application Enablement Services

The Avaya Aura<sup>TM</sup> Application Enablement Services server enables Computer Telephony Interface (CTI) applications to control and monitor telephony resources on Communication Manager. The AES server receives requests from CTI applications, and forwards them to Communication Manager. Conversely, the AES server receives responses and events from Communication Manager and forwards them to the appropriate CTI applications.

This section assumes that installation and basic administration of the AES server has been performed. The steps in this section describe the configuration of a Switch Connection, a CTI user, a CMAPI port, and creating a CTI link for TSAPI.

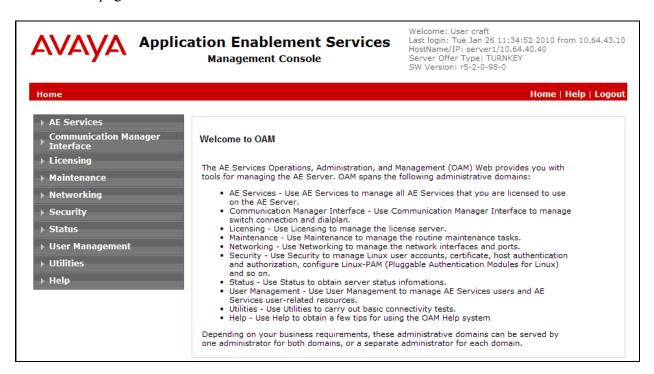
## 5.1. Configure Switch Connection

Launch a web browser, enter https://<IP address of AES server> in the address field, and log in with the appropriate credentials for accessing the AES CTI OAM pages.



Click on Communication Manager 

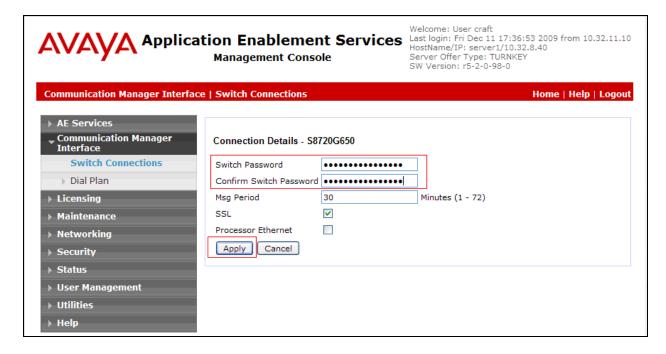
Switch Connection in the left pane to invoke the Switch Connections page.



A Switch Connection defines a connection between the AESs server and Communication Manager. Enter a descriptive name for the switch connection and click on **Add Connection**.



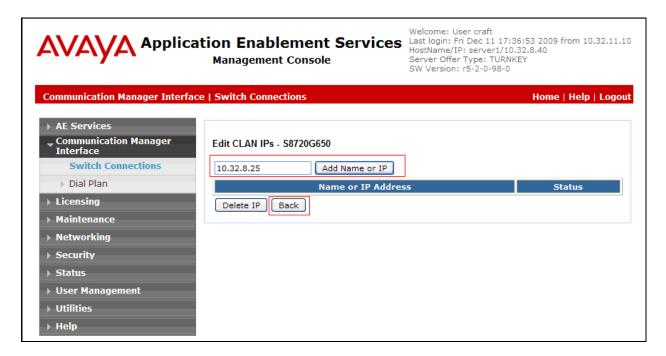
The next window that appears prompts for the Switch Connection password. Enter the same password that was administered in Communication Manager in **Section 4.3**. Click on **Apply**.



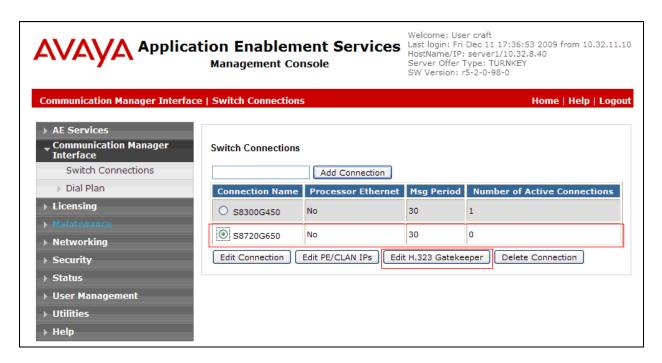
After returning to the Switch Connections page, select the radio button corresponding to the switch connection added previously, and click on **Edit PE/CLAN IPs**.



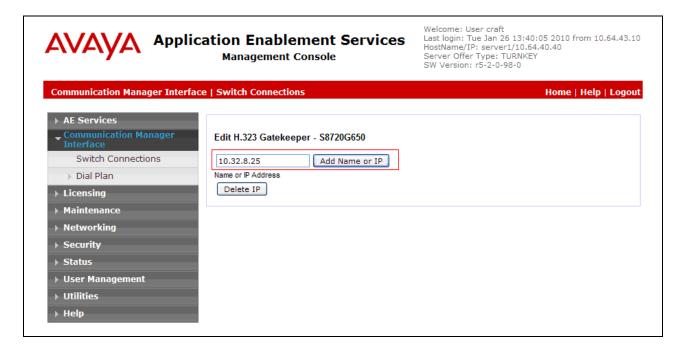
Enter the CLAN-AES IP address which was configured for AES connectivity in **Section 4.3** and click on **Add Name or IP**. Repeat this step as necessary to add other C-LAN boards enabled with Application Enablement Services. Click on the **Back** button after the completion.



On the Switch Connections page, click on **Edit H.323 Gatekeeper** for DMCC call control and monitor.

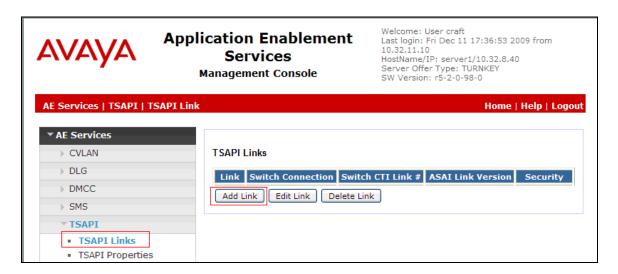


On the **Edit H.323 Gatekeeper – S8720G650** page, enter the C-LAN IP address which will be used for the DMCC service. Click on **Add Name or IP**. Repeat this step as necessary to add other C-LAN boards enabled with Application Enablement Services.



## 5.2. Configure the TSAPI CTI link

Navigate to **AE Services** → **TSAPI** → **TSAPI Links** in the left pane, and click on the **Add Link** button to create a TSAPI CTI link.



Select a Switch Connection using the drop-down menu. The Switch Connection is configured in **Section 5.1**. Select the Switch CTI Link Number using the drop-down menu. Switch CTI Link Number should match with the number configured in the cti-link form **in Section 4.3**. Click the **Apply Changes** button. Default values may be used in the remaining fields.



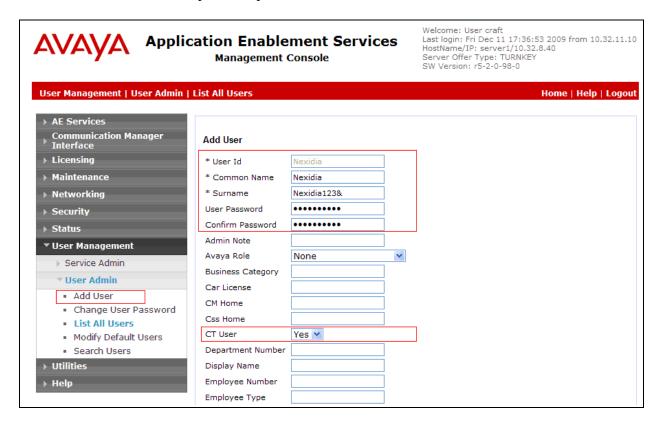
## 5.3. Configure the CTI Users

Navigate to **User Management** → **User Admin** → **Add User**. On the Add User page, provide the following information:

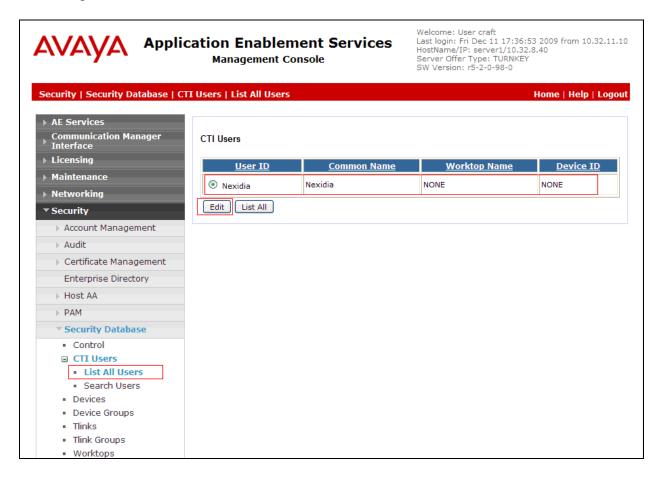
- User Id
- Common Name
- Surname
- User Password
- Confirm Password

The above information (User ID and User Password) must match with the information configured in the ESI-Capture Configuration page in **Section 6**.

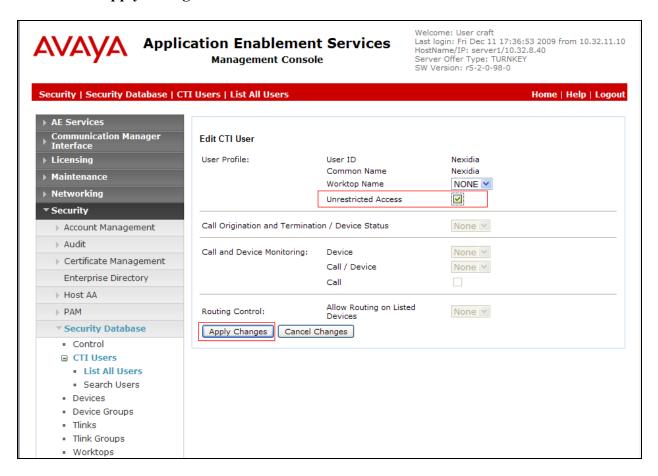
Select **Yes** using the drop down menu on the CT User field. This enables the user as a CTI user. Default values may be used in the remaining fields. Click the **Apply** button (not shown) at the bottom of the screen to complete the process.



Once the user is created, navigate to the CTI OAM Security  $\rightarrow$  Security Database  $\rightarrow$  CTI Users  $\rightarrow$  List All Users page. Select the User ID created previously, and click the Edit button to set the permission of the user.

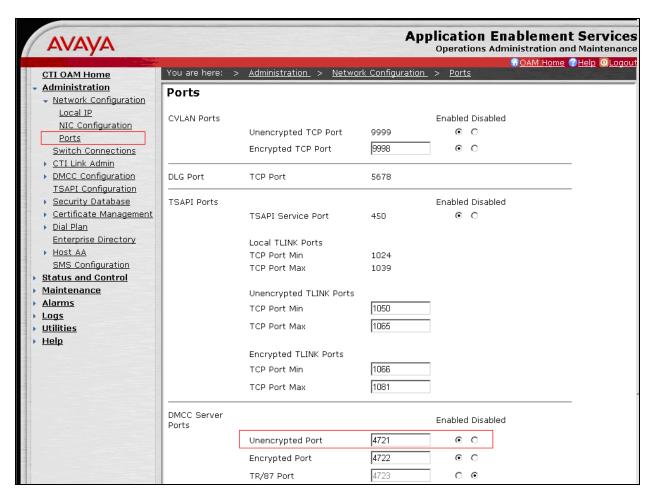


Provide the user with unrestricted access privileges by checking the **Unrestricted Access** button. Click on the **Apply Changes** button.



## 5.4. Configure the CTI Port

Navigate to the **Administration** → **Network Configuration** → **Ports** page to set the DMCC server port. During the compliance test, the default port values were utilized. The following screen displays the default port values. Since the unencrypted port was utilized during the compliance test, set the Unencrypted Port field to **Enabled**. Default values may be used in the remaining fields. Click the **Apply Changes** button (not shown) at the bottom of the screen to complete the process.



## 6. Configure Nexidia ESI-Capture

This section only describes the interface configuration for the ESI-Capture application to communicate with AES and Communication Manager.

Refer to **Section 10, [3] and [4]** for configuring the ESI-Capture application. The following screen shows the global properties file. During the compliance test, the highlighted values were utilized:

```
#nx.switch.name=cmsim
nx.switch.name=S8720G650
nx.capture.channel0.extension=23001
nx.capture.channel0.password=1234
nx.capture.channel1.extension=23002
nx.capture.channel1.password=1234
nx.capture.channel2.extension=23003
nx.capture.channel2.password=1234
nx.capture.channel3.extension=23004
nx.capture.channel3.password=1234
nx.capture.channel4.extension=23005
nx.capture.channel4.password=1234
#nx.capture.channel5.extension=32205
#nx.capture.channel6.extension=32206
#nx.capture.channel7.extension=32207
#nx.capture.channel8.extension=32208
#nx.capture.channel9.extension=32209
nx.manager.monitor0.extension=22001
nx.manager.monitor1.extension=22002
nx.manager.monitor2.extension=22003
nx.manager.monitor3.extension=22007
nx.manager.monitor4.extension=22009
#nx.manager.monitor4.extension=32404
#nx.manager.monitor5.extension=32405
#nx.manager.monitor6.extension=32406
#nx.manager.monitor7.extension=32407
#nx.manager.monitor8.extension=32408
#nx.manager.monitor9.extension=32409
nx.capture.codecs=g711U
#cmapi.server_ip=192.168.1.224
cmapi.server ip=10.32.11.40
cmapi.server port=4721
#cmapi.username=craft
#cmapi.password=craft01
cmapi.username=Nexidia
cmapi.password=Nexidia123&
# It is important to set this value in production to enable session recovery
without
# reestablishing state(reregistering the stations)
cmapi.session cleanup delay=60
```

## 7. General Test Approach and Test Results

The general approach was to manually place calls to and from stations, monitor and record them using the ESI-Capture, and verify the recordings. The types of calls included internal calls, inbound, outbound trunk calls, transfer calls, conference calls. For serviceability testing, failures such as cable pulls, CTI link busyouts and releases, and resets were applied.

The test objectives were verified. For serviceability testing, Nexidia ESI-Capture operated properly after recovering from failures such as cable disconnects, and resets of Nexidia ESI-Capture and the SIP Enablement Services server.

## 8. Verification Steps

This section provides the steps that can be performed to verify proper configuration of Communication Manager and AES.

## 8.1. Verify Communication Manager

Verify the status of the administered AES link by using the **status aesvcs link** command.

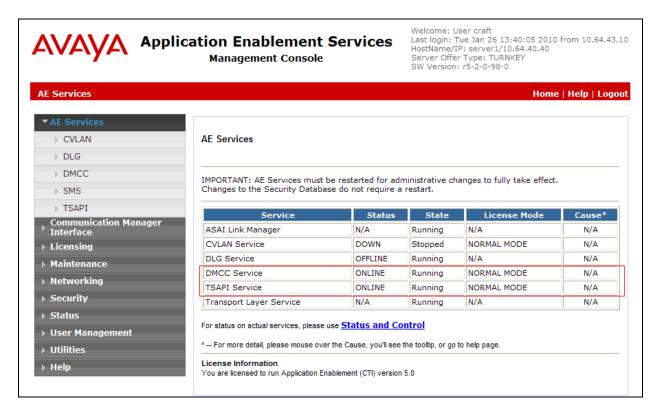
status	aesvcs link				
		AE SERVICES	LINK STATUS		
Srvr/ Link	AE Services Server	Remote IP	Remote Local Node Port	Msgs Sent	Msgs Rcvd
01/01	server2	10.32.8.40 6033	6 CLAN-AES 208	197	

Verify the Service State field of the administered TSAPI CTI link is in **established** state, by using the **status aesvcs cti-link** command.

statu	s aesvcs	cti-li	nk			
			AE SERVICES	CTI LINK STAT	TUS	
CTI Link	Version	Mnt Busy	AE Services Server	Service State	Msgs Sent	Msgs Rcvd
4	4	no	server2	established	15	15

## 8.2. Verify Avaya Application Enablement Services

From the CTI OAM Admin web pages, verify the status of the TSAPI and DMCC Services are ONLINE, by selecting **AE Services** from the left pane.



## 9. Conclusion

These Application Notes illustrate the procedures for configuring the ESI-Capture call recording solution to monitor and record calls placed to and from stations on an Communication Manager system. In the configuration described in these Application Notes, the ESI-Capture employs Device, Media and Call Control Application Programming Interface virtual stations as recording ports. During compliance testing, the ESI-Capture successfully monitored events and recorded calls placed to and from stations.

## 10. Additional References

This section references the Avaya and Nexidia documentation that are relevant to these Application Notes.

The following Avaya product documentation can be found at <a href="http://support.avaya.com">http://support.avaya.com</a>.

- [1] *Administering Avaya Aura* TM *Communication Manager*, Issue 5.0, May 2009, Document Number 03-300509
- [2] Application Enablement Services Administration and Maintenance Guide, Release 5.2, Issue 11, November 2009, Document Number 02-300357

The following documentation was provided by Nexidia

- [3] Nexidia ESI--Capture Install and Config Guide, Issue 1.0, October, 2009
- [4] Nexidia ESI--Capture System Requirements, Issue 1.0

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