

#### Avaya Solution & Interoperability Test Lab

Application Notes for Configuring NICE Interaction Management R4.1 with Avaya Aura® Contact Centre R6.2 and Avaya Aura® Application Enablement Services R6.1 for Call Recording in a Mission Critical High Availability Environment – Issue 1.0

#### **Abstract**

These Application Notes describe the configuration steps for provisioning NICE Interaction Management R4.1 with a SIP enabled Avaya Aura® Contact Centre R6.2 in a full High Availability Mission Critical environment for call recording. NICE Interaction Management records the RTP stream coming from the Avaya Media Server module of Avaya Aura® Contact Centre using events from the Communication Control Toolkit module of Avaya Aura® Contact Centre.

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 the compliance tested configuration required for interoperability between NICE Interaction Management R4.1 and Avaya Aura® Contact Centre R6.2 in a Mission Critical High Availability environment. To achieve the highest level of Mission Critical High Availability with no single point of failure a SIP-based contact centre with the following criteria must be setup. (See **Section 3**, **Figure 1** for a diagram setup of the complete solution).

- Two co-resident Contact Centre Manager Server (CCMS), Communication Control Toolkit (CCT), and Contact Centre Manager Administration (CCMA) servers configured as a High Availability pair.
- Two or more Avaya Media Server Linux-based servers configured as a High Availability pair. Avaya Media Server High Availability is supported only on Linux-based servers.
- Two Avaya Aura® Session Manager instances, R6.1.
- Two Avaya Aura® Application Enablement Services servers configured as a High Availability pair.
- Two Avaya Aura® Communication Manager Servers configured as a High Availability pair.

NICE Interaction Management R4.1 is a software-only solution that offers various recording, playback and archiving features and options. By combining media redirection from Avaya Aura® Contact Centre, call recording can be achieved without the use of physical connections to the NICE server other than standard network connections. The NICE solution is fully integrated into a LAN (Local Area Network), and includes Web based applications (i.e. NICE Applications) that work with .NET framework that are used to retrieve telephone conversations from a database of recorded voice calls. These Application Notes focus on recording calls from agents on a skillset call. NICE Interaction Management's internal scheduling algorithm makes the determination on which calls should be recorded based on the events received from Web Services on the Communication Control Toolkit module of Avaya Aura® Contact Centre.

# 2. General Test Approach and Test Results

The compliance testing focuses on the recording of Avaya Aura® Contact Centre (Contact Centre) skillset calls on Communication Manager deskphones. NICE Interaction Management connects to Communication Control Toolkit (CCT) Web Services in order to obtain events pertaining to specific Contact Centre skillset calls. Interaction Management can then record the call based on the events it receives. When a call is to be recorded, the Interaction Management performs recording using CCT web service recording API to enable SIP recording with Avaya Media Server. Avaya Media Server then forwards all RTP packages to NICE SIP Logger.

In a High Availability Environment one set of Contact Centre applications, a CCMS, a CCT and a CCMA actively processes scripts and contacts. This set of applications is called the active set. Another set of Contact Centre applications in the same Contact Centre system monitors and shadows the active applications in the system. The standby applications track the state of active calls but do not process calls. The standby CCMS monitors the active CCMS. The standby CCT monitors the active CCT. Each active and standby pair of applications forms a resilient or replication pair. If any of the active applications fail, the standby applications recognize the failure and start processing contacts. Contact Centre Administrators use the active server in daily operation. Configuration changes made to the active system during normal operation are automatically copied to the standby applications, therefore the standby applications are configured and ready to take over processing from the active system. Statistical data is also automatically copied to the standby applications. Data is replicated to the standby applications in real time. When the Contact Centre fails over to the standby server a new socket connection must be made between the NICE Interaction Management and the Standby CCT Web Services. Please see Section 2.2 for observations during the failover testing.

Recording of Contact Centre skillset calls is done using CCT web services, all other calls use Device Media and Call Control (DMCC) to perform service observe between the extension to be recorded and a configured virtual softphone enabled station. The recording application sends a message to the DMCC integration application to begin recording the voice stream coming to that softphone extension. NICE Interaction Management utilises a CTI through Avaya Aura® Application Enablement Services (AES) to record calls on Communication Manager deskphones using Service Observe. In this message, the recorder passes along the softphone extension to be recorded along with the location and filename of the recording. Test cases are executed to exercise a sufficiently broad segment of functionality to have a reasonable expectation of interoperability in production configurations.

DevConnect Compliance Testing is conducted jointly by Avaya and DevConnect members. The jointly-defined test plan focuses on exercising APIs and/or standards-based interfaces pertinent to the interoperability of the tested products and their functionalities. DevConnect Compliance Testing is not intended to substitute full product performance or feature testing performed by DevConnect members, nor is it to be construed as an endorsement by Avaya of the suitability or completeness of a DevConnect member's solution.

#### 2.1. Interoperability Compliance Testing

The testing focuses on the following types of calls:

- Communication Manager Inbound/Outbound calls Test call recording for inbound/outbound calls to the Communication Manager from PSTN callers.
- Communication Manager Hold/Transferred/Conference calls Test call recording for calls transferred to and in conference with PSTN callers.
- Contact Centre Inbound/Outbound Calls Test call recording for inbound/outbound calls to the Contact Centre Agents from PSTN callers.
- Contact Centre Hold/Transferred/Conference calls Test call recording for calls transferred to and in conference with PSTN callers.
- Contact Centre Record on demand/Stop on demand (ROD/SOD) allow agents to stop and start recordings during a telephone conversation.
- HA Failover from Contact Centre active to Contact Centre standby The behaviour
  of NICE Recording Solution under different simulated LAN failure conditions on the
  Avaya Contact Centre Active Server.
- HA Failover from Communication Server active to Communication Server standby

   The behaviour of NICE Recording Solution under different simulated LAN failure conditions on Communication Server.
- HA Failover from Media Server active to Media Server standby The behaviour of NICE Recording Solution under different simulated LAN failure conditions on the Media Server
- **HA Failover from AES active to AES standby** The behaviour of NICE Recording Solution under different simulated LAN failure conditions on AES.

#### 2.2. Test Results

All compliance test cases passed successfully. There were no errors observed on the Avaya Solution as a result of the addition of NICE Integration Management to the LAN. The following observations were noted during the failover testing of the Contact Centre from Active to standby.

- The Contact Centre failure from Active to Standby can occur in less than 3 seconds depending on the failure that occurs on the Active server.
- Web Services on the CCT module does not broadcast a terminate or failover message to the connected sessions.
- Timers on the NICE Interaction Management for a keep-alive message to Web Services on the Contact Centre CCT module were adjusted during the HA Failover from Contact Centre active to Contact Centre standby. This was to facilitate an issue were the CCT was failed over but the NICE interaction Management was not aware and was connected to an expired session on Web Services. In order to ensure that a new session was established on all occasions the NICE Interaction Management needed to send and receive a keep-alive message to Web Services every second to ensure it was aware when the active session was stopped and a switch over occurred.

#### 2.3. Support

Support from Avaya is available at <a href="http://support.avaya.com">http://support.avaya.com</a> and support from NICE can be obtained as shown below.

NICE International Corporate Headquarters, Israel

Tel: +972 9 775 3800 Email: support@nice.com

# 3. Reference Configuration

**Figure 1** shows the compliance tested configuration which includes duplicate Communication Manager servers in High Availability, Session Manager to provide SIP functionality, AES to provide DMCC events from Communication Manager and Contact Centre which includes the CCT module to provide call events for Contact Centre calls and Media Server to provide the RTP for recording.

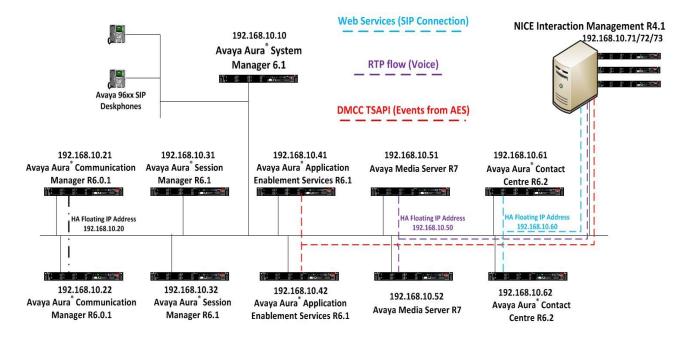


Figure 1: Connection of NICE Interaction Management R4.1 for interoperability with Avaya Aura® Contact Centre R6.2 and Avaya Aura® Application Enablement Services R6.1 in a Mission Critical High Availability Environment.

# 4. Equipment and Software Validated

The following equipment and software were used for the sample configuration provided.

Equipment/Software	Release/Version
Avaya Aura® System Manager running on Avaya S8800 Server	R6.1 SP4
Avaya Aura® Communication Manager running on Avaya S8800 Server	R6.0.1 SP3 (Active Server HA mode)
Avaya Aura® Communication Manager running on Avaya S8800 Server	R6.0.1 SP3 (Standby Server HA mode)
Avaya Aura® Session Manager running on Avaya S8800 Server	R6.1 SP4 (Primary -Active Server in Active-Active Mode)
Avaya Aura® Session Manager running on Avaya S8800 Server	R6.1 SP4 (Secondary -Active Server in Active-Active Mode)
Avaya Aura® Application Enablement Services running on Avaya S8800 Server	R6.1 (Active Server with System Platform in HA)
Avaya Aura® Application Enablement Services running on Avaya S8800 Server	R6.1 (Standby Server with System Platform in HA)
Avaya Aura® Contact Centre running on Avaya S8800 Server	R6.2 SIP enabled(Active Server in HA Mode)
Avaya Aura® Contact Centre running on Avaya S8800 Server	R6.2 SIP enabled(Standby Server in HA Mode)
Avaya Media Server running on Avaya S8800 Server	R7 running on Redhat Linux R5.4 (Active Sever)
Avaya Media Server running on Avaya S8800 Server	R7 running on Redhat Linux R5.4 (Standby Sever)
Avaya 96xx Series Deskphone	96xx H.323 Release 3.1 SP2
Avaya 96xx Series Deskphone	96xx SIP Release 2.6 SP3
NICE Interaction Management Server running Windows 2008 O/S	NICE Interaction Management 4.1 Update Pack 22
NICE Interaction Management Server running Windows 2008 O/S	NICE VoIP SIP Logger 4.1 Update Pack 22
NICE Interaction Management Server running Windows 2008 O/S	NICE VoIP DMCC Logger 4.1 Update Pack 22

# 5. Configure Avaya Aura® Communication Manager

The setup of Communication Manger in a High Availability environment is outside the scope of these Application Notes. It is therefore assumed that a fully functioning High Availability Communication Manager is in place with the necessary licensing and a SIP connection is already made to Session Manager. For further information on the configuration of Communication Manager please see **Section 12** of these Application Notes.

#### 5.1. Configure TSAPI CTI Link

Enter the **add cti-link x** command, where **x** is a number between 1 and 64, inclusive. Enter a valid **Extension** under the provisioned dial plan. Set the **Type** field to **ADJ-IP** and assign a descriptive **Name** to the CTI link. Default values may be used in the remaining fields.

```
add cti-link 1

CTI Link: 1

Extension: 2100

Type: ADJ-IP

Name: AACC
```

Enter the **change node-names ip** command. In the compliance-tested configuration, the **procr** IP address was utilized for registering H.323 endpoints and connectivity to the Application Enablement Services server. Note also the AES server name and IP address added, **AES61** and IP Address **192.168.10.41**.

change node-names	ip	Page	1 of	2
	IP NODE NAMES			
Name	IP Address			
AES61	192.168.10.41			
SM100-1	192.168.10.31			
SM100-2	192.168.10.32			
clan	192.168.10.102			
default	0.0.0.0			
gateway	192.168.10.1			
medpro	192.168.10.103			
procr	192.168.10.20			
procr6	::			

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 **procr** that was configured previously in the node-name ip form. During the compliance test, the default port was utilized for the **Local Port** field.

change ip-s	services				Page	<b>1</b> of	3
Service Type	Enabled	Local Node	IP SERVICES Local Port	Remote Node	Remote Port		
AESVCS	У	procr	8765				

On **Page 3**, enter the hostname of the AES server for the **AE Services Server** field. Enter an alphanumeric password for the **Password** field. Set the **Enabled** field to y. The same password will be configured on the Application Enablement Services in **Section 6.1**.

change ip-services  AE Services Administration				<b>3</b> of	3
THE BETVICES HAMITHISETERS					
	Services Password Server	Enabled	Status		
1: aes1	Manchestercity12	У	in use		
2:					

#### 5.2. Configure Virtual Stations for Service Observe

Add virtual stations to allow Interaction Management to record calls using Service Observe. Type **add station x** where x is the extension number of the station to be configured. Also note this extension number for configuration required in **Section 9.1.** Note the **Security Code** and ensure that **IP SoftPhone** is set to y. Note the **COR** of the stations below.

display station 52001		STATION	Page	1 of	5	
Extension: 52001		Lock Messages? n		BCC:	0	
Type: 4621		Security Code: 1234		TN:	1	
Port: S00034		Coverage Path 1:		COR:	1	
Name: Nice VE3		Coverage Path 2:		cos:	1	
		Hunt-to Station:				
STATION OPTIONS						
Location:		Time of Day Lock Table:				
Loss Group:	19	Personalized Ringing Pattern:	1			
		Message Lamp Ext:	5200	1		
Speakerphone:	2-way	Mute Button Enabled?	У			
Display Language:	english	Expansion Module?	n			
Survivable GK Node Name:						
Survivable COR:	internal	Media Complex Ext:				
Survivable Trunk Dest?	У	IP SoftPhone?	, У			
	IP Video Softphone?					
	Short/Prefixed Registration Allowed: default					
		Customizable Labels?	У			

Type **display cor x**, where **x** is the COR number in the screen above, to check the existing Class of Restriction. Ensure that **Can be Service Observed** is set to **y**. If not type **change cor 1** to make a change to Class or Restriction (cor) 1. This needs to be enabled for Service Observe to work properly.

```
display cor 1
                                                                                       Page 1 of 23
                                       CLASS OF RESTRICTION
                    COR Number: 1
             COR Description:
          APLT? y

E service Observed? y

Calling Party Restriction: all-toll

A Service Observer? y

Called Party Restriction: none

Time of Day Chart: 1

Priority Queuing? n

Striction Overside
  Can Be Service Observed? y
Can Be A Service Observer? y
      Restricted Call List: 1

Titority Queuing? n

Direct Agent Calling? y

Facility Access Trunk Test? n

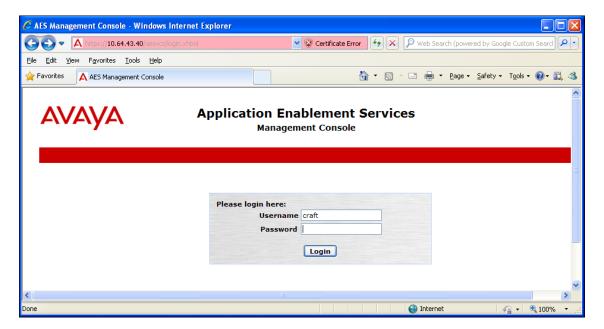
Can Change Covers
    Unrestricted Call List: 1
Access to MCT? y
Group II Category For MFC: 7
Send ANI for MFE? n
MF ANI Prefix:
                                                  Fully Restricted Service? n
                                                    Hear VDN of Origin Annc.? n
                                                     Add/Remove Agent Skills? n
                                                    Automatic Charge Display? n
Hear System Music on Hold? y PASTE (Display PBX Data on Phone)? n
                                Can Be Picked Up By Directed Call Pickup? y
                                                Can Use Directed Call Pickup? y
                                                Group Controlled Restriction: inactive
```

# 6. Configure Avaya Aura® Application Enablement Services

Application Enablement Services enable Computer Telephony Interface (CTI) applications to control and monitor telephony resources on Communication Manager. Application Enablement Services (AES) receives requests from CTI applications, and forwards them to Communication Manager. Conversely, Application Enablement Services (AES) receives responses and events from Communication Manager and forwards them to the appropriate CTI applications.

**Note:** The installation and setup of the AES in a High Availability environment is outside the scope of these Application Notes and it is therefore assumed that installation and basic administration of the Application Enablement Services server has been performed. The steps in this section describe the configuration of a Switch Connection, creating a CTI link for TSAPI, and a CTI user. For further information on Avaya Application Enablement Services please refer to **Section 12** of these Application Notes.

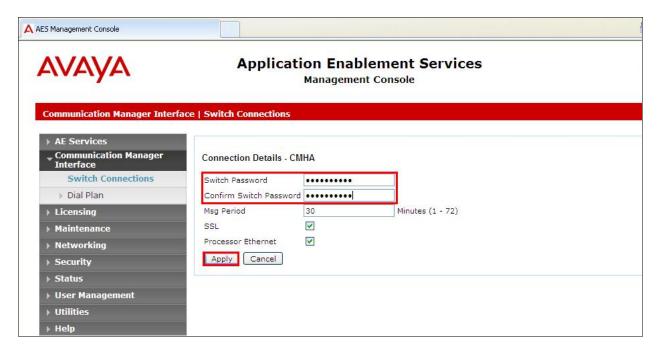
Launch a web browser, enter https://<IP address of AES server> in the URL, and log in with the appropriate credentials for accessing the Application Enablement Services Management Console page.



Click on Communication Manager Interface  $\rightarrow$  Switch Connections in the left pane to invoke the Switch Connections page. A Switch Connection defines a connection between the Application Enablement Services 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 Password. Enter the same password that was administered on Communication Manager in **Section 5.1**. Default values may be used in the remaining fields. 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**.

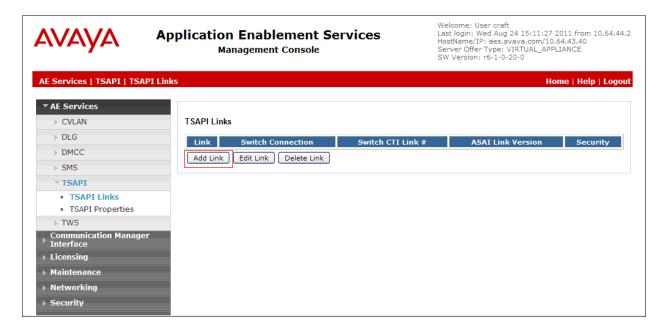


Enter the IP address of the procr from Section 5.1, and click on Add/Edit Name or IP.

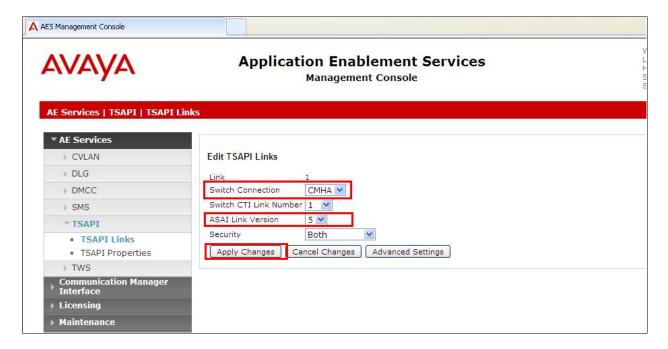


#### 6.1. Configure TSAPI CTI Link

Navigate to **AE Services** → **TSAPI** → **TSAPI Links** to configure the TSAPI CTI link. Click the **Add Link** button to start configuring the TSAPI link.



Select the switch connection using the drop-down menu. Select the switch connection configured in **Section 6.1**. Select the **Switch CTI Link Number** using the drop-down menu. The **ASAI Link Version** is set to **5**. The CTI link number should match with the number configured in the CTI-link in **Section 5.1**. Click **Apply Changes**.

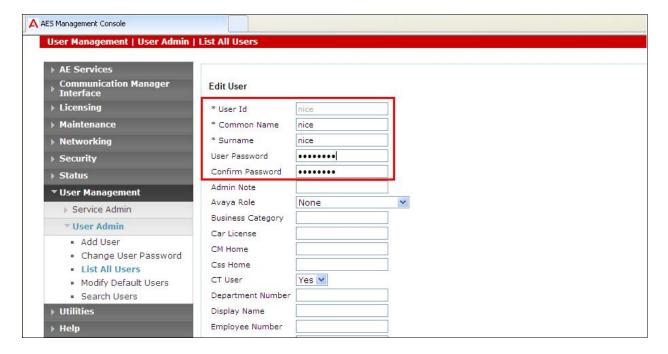


#### 6.2. Configure CTI User

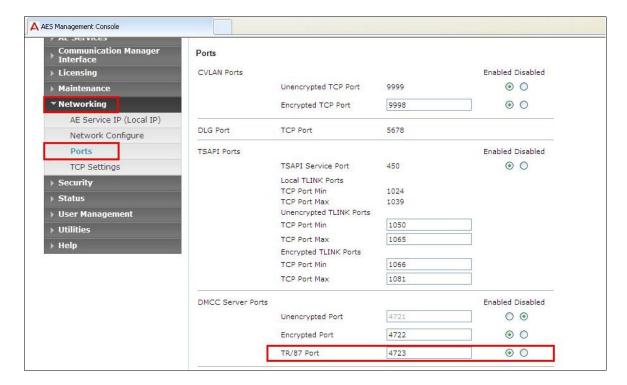
Navigate to **User Management** → **Add User**. On the **Add User** page, provide the following information.

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

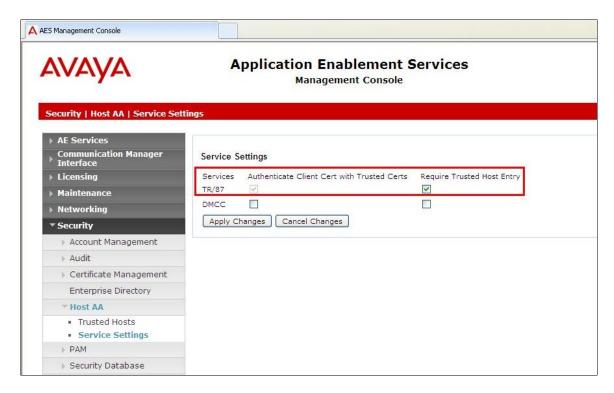
Select **Yes** using the drop-down menu on the **CT User** field. This enables the user as a CTI user. Click the **Apply** button (not shown here) at the bottom of the screen to complete the process. Default values may be used in the remaining fields.



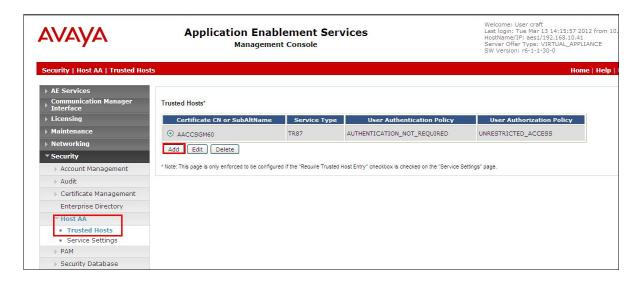
Click on **Networking** → **Ports** enable port 4723 for the **TR87** SIP Interface as shown below.



Click on Security → Host AA → Service Settings. Ensure that Require Trusted Host Entry is ticked.



Click on Security → Host AA → Trusted Hosts, click on Add to add a new trusted host.



Enter the information as it is shown below. Ensure **Service Type** is **TR/87** and click on **Apply Changes**.



# 6.3. Avaya Aura® Application Enablement Services Certificate Management

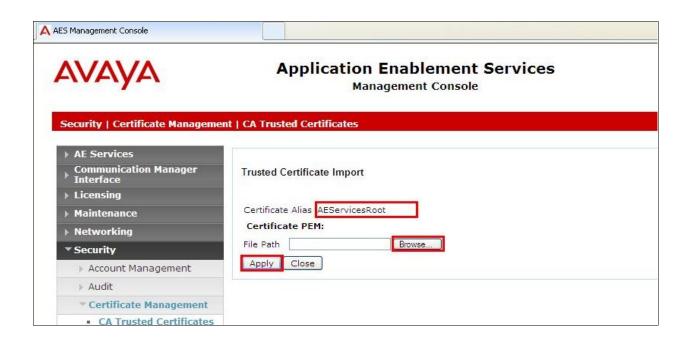
Click on Security  $\rightarrow$  Certificate Management  $\rightarrow$  CA Trusted Certificates. Click on Import to import the Certificates from Contact Centre. The certificates required and their location are outlined in Section 7.3 of these Application Notes. These certificates are as follows.

- AACCSGM60Root.pem
- AEServicesRoot.cer

**Note:** An avi video outlined in **Section 7.3** is available giving instructions on adding the certificates onto AES.

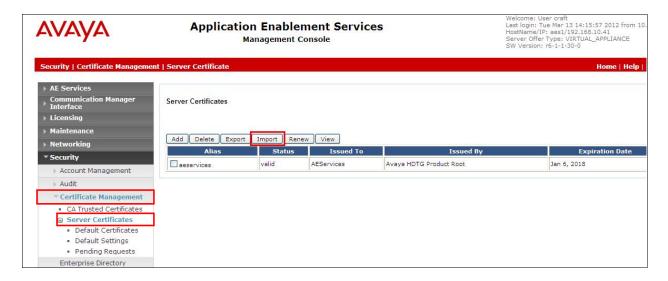


Once imported is selected above the following screen appears. Below is an example of adding the **AEServicesRoot** certificate. Browse to the location of this certificate and once selected (not shown) click **Apply**.

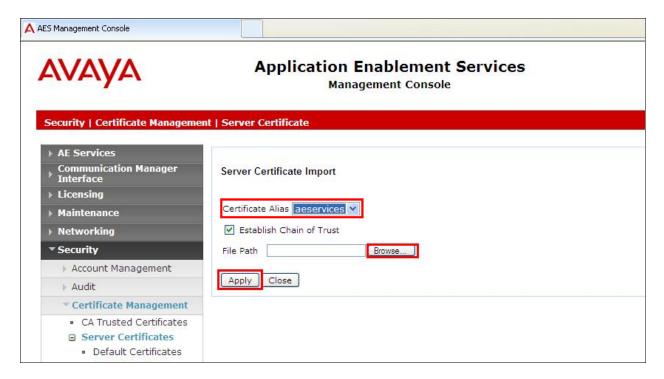


Once the two **CA Trusted Certificates** are added, click on **Server Certificates**, click on Import to import the server certificate from Contact Centre. The location of this certificate is shown in **Section 7.3** of these Application Notes. This certificate is as follows.

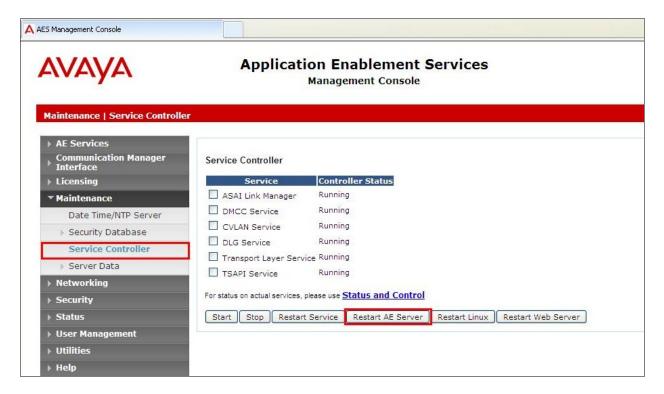
AEServices



Once **Import** is selected above the following screen appears. Select **aeservices** from the drop-down menu, click on **Browse** to locate the certificate, once selected (not shown) click **Apply**.



Once all the certificates are added the services need to be restarted. Click **Maintenance** > **Service Controller**. Click on **Restart AE Server** as highlighted below to restart all the services.



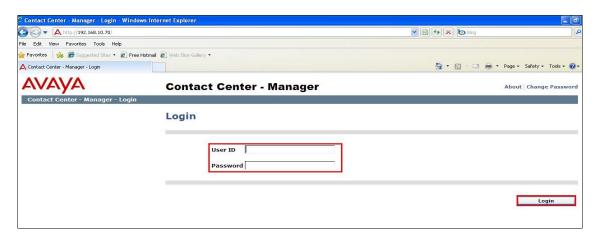
# Configure Avaya Aura® Contact Centre

In order for NICE Interaction Management to be able to record calls from Contact Centre Agent calls a user must be configured on Contact Centre in CCT. This user can then log in to see events from CCT regarding the calls to Contact Centre agents. This section will go through the setup of this agent and the configuration necessary on both CCT and web services in order to record all calls coming into Avaya Contact Centre agents.

**Note:** The Installation and Setup of Contact Centre in a Mission Critical High Availability environment is outside the scope of these Application Notes. This section assumes that installation and basic administration of the Contact Centre server has been performed. The steps in this section describe the configuration of Contact Centre in order for NICE interaction Management to connect to CCT to receive events and successfully receive RTP from Avaya Media Server. For further information on Contact Centre please refer to **Section 12** of these Application Notes.

### 7.1. Configure NICE User on CCT

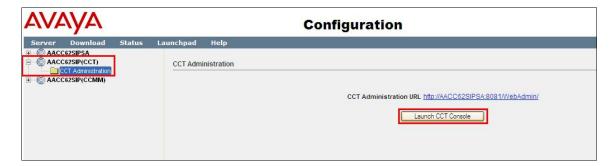
Launch a web browser, enter https://<IP address of Contact Centre server> in the URL, and log in with the appropriate credentials for accessing the Contact Center - Manager Console page.



Once logged in, click on **Configuration** as shown below.



Expand on the CCT server on the left-hand pane as shown and select CCT Administration. Click on Launch CCT Console in the right-hand pane.



Right click on Uses highlighted below.



Click on Add New User.

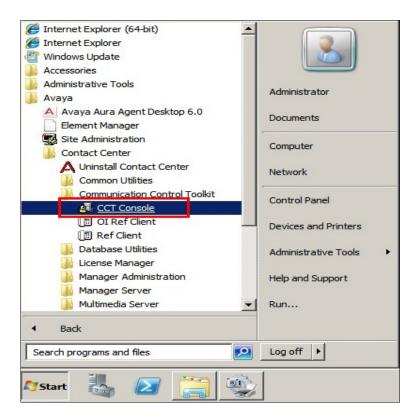


Enter the credentials as shown below. Note these same credentials will be used in **Section 9.1** of these Application Notes.

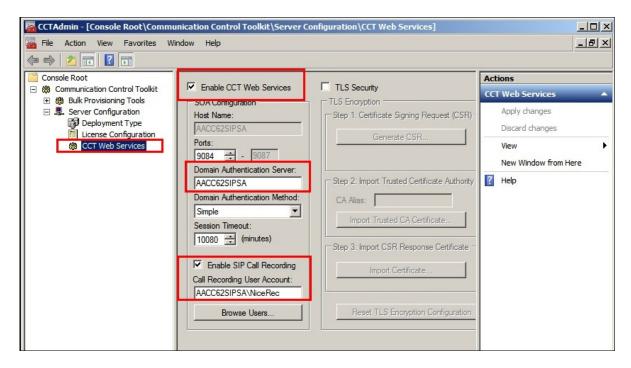


# 7.2. Enabling CCT Web Services

On the Contact Centre server navigate to **Start**  $\rightarrow$  **Programs**  $\rightarrow$  **Avaya**  $\rightarrow$  **CCT Console** as shown below.

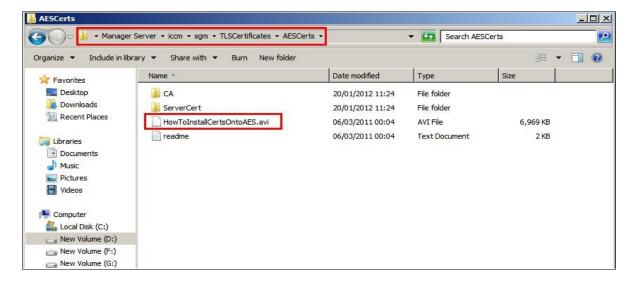


The CCTAdmin page is displayed as shown. Click on CCT Web Services in the left column and ensure that Enable CCT Web Services is ticked along with Enable SIP Call Recording. Enter the Contact Centre server name for Domain Authentication Server and the user configured in Section 7.1 for the Call Recording User Account.

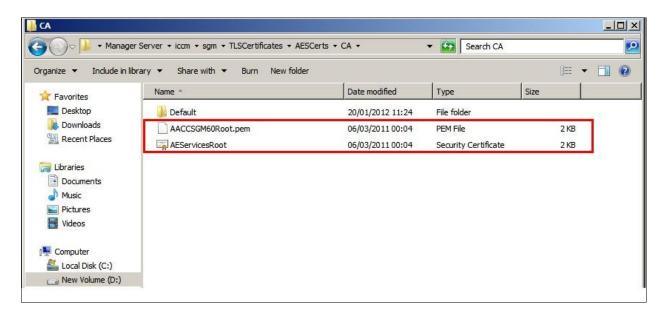


# 7.3. Locating Certificates for Avaya Aura® Application Enablement Services

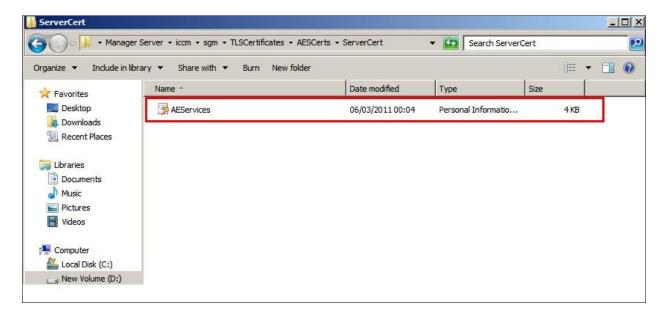
Locate the certificates required for AES by navigating to D: → Avaya → Manager Server → iccm → sgm → TLSCertificates → AESCerts. Two CA Trusted Certificates are located in the CA folder and the Server certificate is located on the ServerCert as shown below. To assist in adding these certificates in Section 6.3 open the avi named HowToInstallCertsOntoAES.



Open the folder above called CA, copy the two files highlighted below to a location for use in **Section 6.3**.

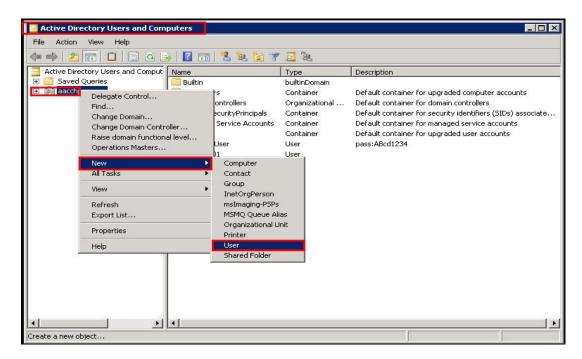


Open the folder **ServerCert** as shown in the previous page, copy the file highlighted below to a location for use in **Section 6.3**.

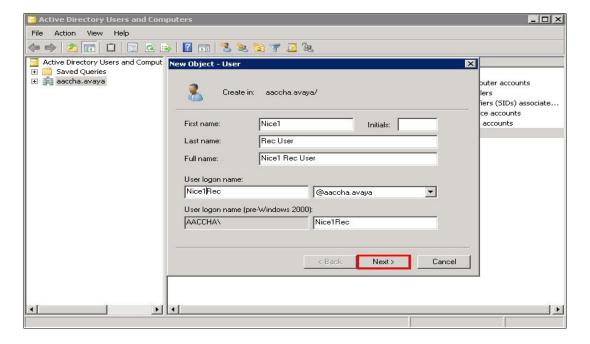


# 8. Configure Domain Controller

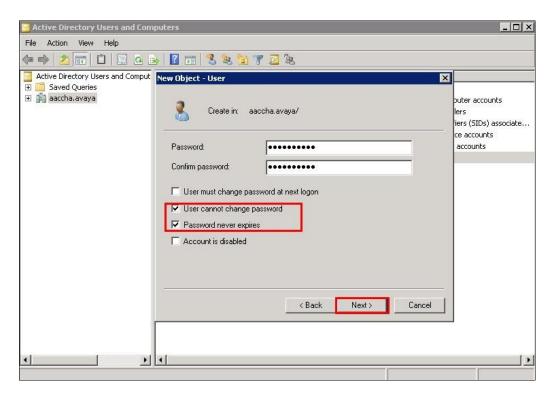
The CCT user configured in **Section 7.1** above must also be configured as a domain user on the Primary Domain Controller Server. Open **Active Directory Users and Computers** and right click on the server name, select **New → User**.



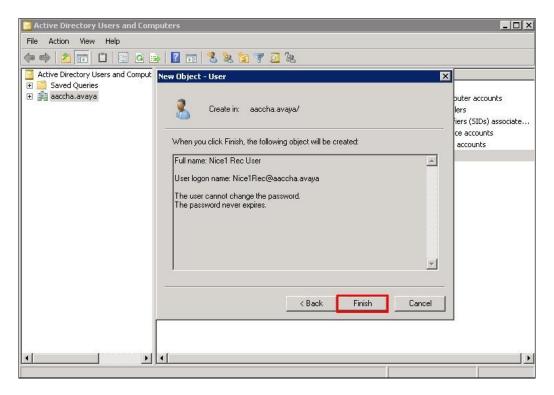
Fill in the User logon name and other credentials as shown, click Next to continue.



Enter a suitable **Password** and ensure that **User cannot change password** and **Password never expires** are ticked. Click **Next** to continue.



Click Finish to complete adding the new user.



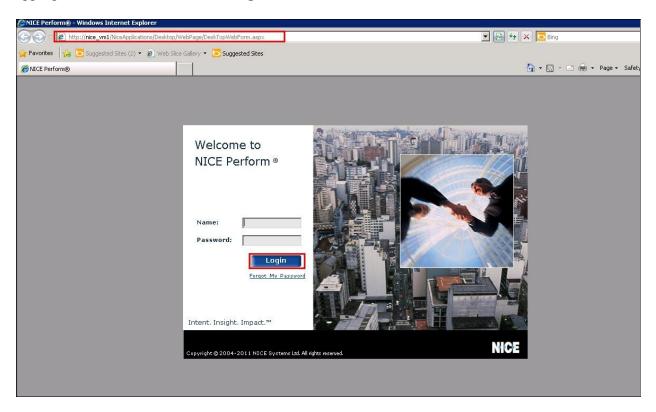
# 9. Configure NICE Integration Management

This section outlines the steps necessary to configure NICE Integration Management to connect successfully to the Avaya Solution outlined in **Section 3** of these Application Notes. The NICE Solution connects to the Communication Control Toolkit (CCT) module of Contact Centre as a CCT agent in order to receive events from the Contact Centre. These events are only passed to the NICE server when a Skillset call is being received by the agent. In order to receive events for calls made on the Avaya Deskphones an interface to the AES is configured to receive events via DMCC and TSAPI.

**Note:** In the case of a High Availability Contact Centre the NICE Interaction Management connects to a floating IP Address. This is common to both the active server and standby server and thus never changes regardless of the system that is active.

# 9.1. Configure NICE Interaction Management to connect to Communication Control Toolkit

Open a web browser, navigate to http://<NICE Interaction machine name>. Enter the appropriate credentials and click Login.

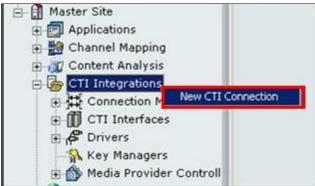


After logging in click on **Settings** highlighted below and choose **Technician Mode**.

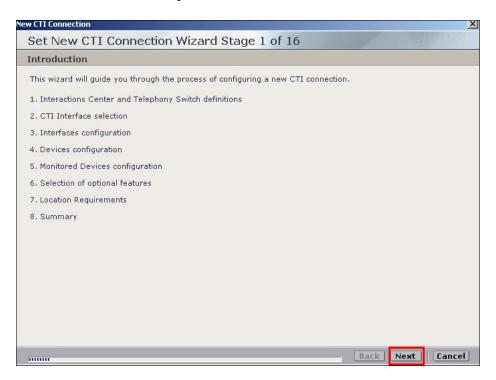


Under **Administration** at the top right select **System Administrator**. Right click on CTI integrations and select **add New CTI Connection** (see below).

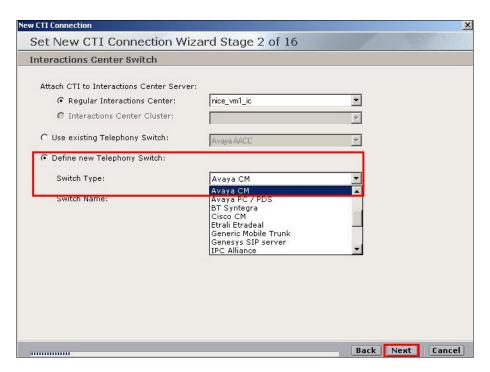




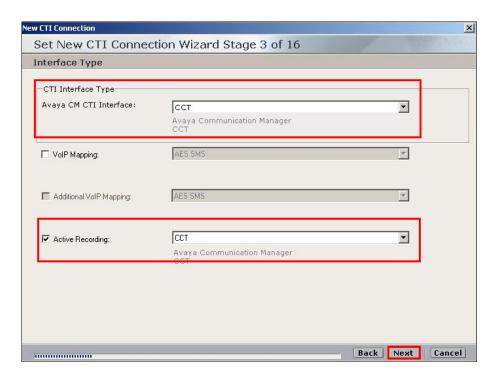
The New CTI Connection window opens as shown below. Click Next to continue.



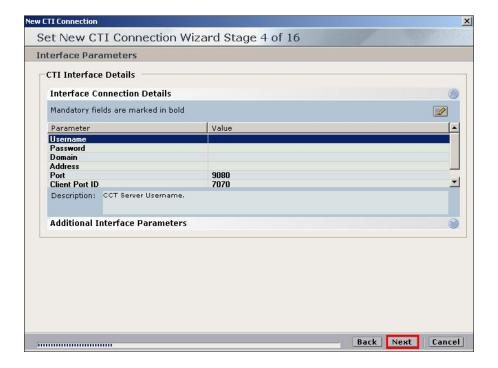
Select **Define new Telephony Switch** and ensure **Avaya CM** is picked from the drop-down menu. Click **Next** to continue.



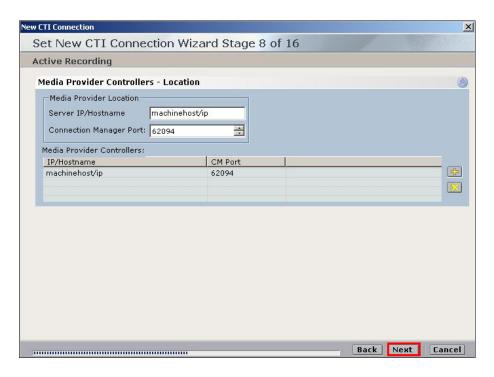
Ensure CCT is chosen for both Avaya CM CTI Interface and Active Recording as shown below. Click Next to continue.



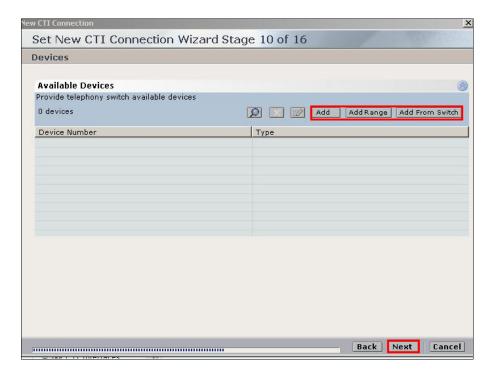
Enter the connection details (Username, Password, Domain and Address) to the CCT as configured in Section 7.1. Click Next to continue.



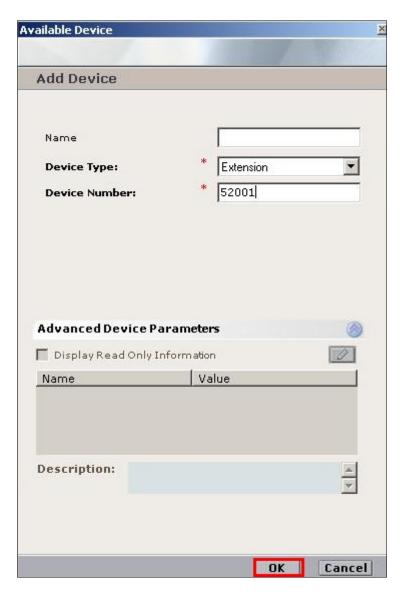
Enter the **Media Provider Controllers – Location**; this will be the IP address of the NICE logger server as shown in **Section 9.3**. Click **Next** to continue.



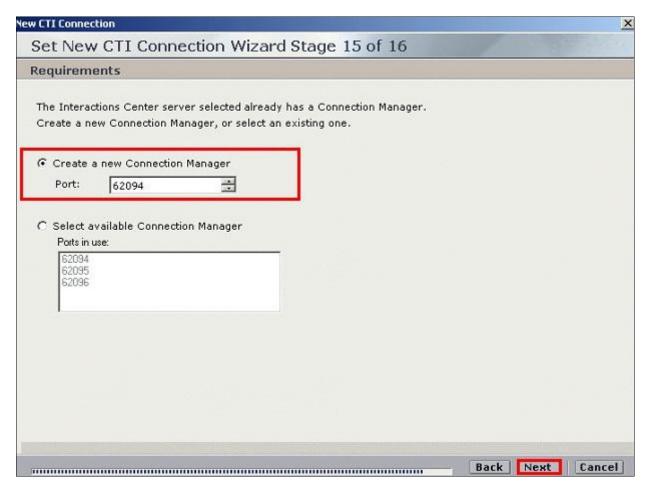
Add **Telephony Switch Devices** in order to record private DN calls via DMCC. Each **Device** or extension can be added singly or in a **Range** of extensions. Click on **Add** highlighted below.



Enter a suitable **Name**, select **Extension** for **Device Type** and enter the extension number for each deskphone that is to be recorded for the **Device Number**. Click **OK** when finished.



Select Create a new Connection Manager and use a unique port. Click Next to pass to the summary window (shown below).





# 9.2. Configure NICE Interaction Management to connect to Application Enablement Services

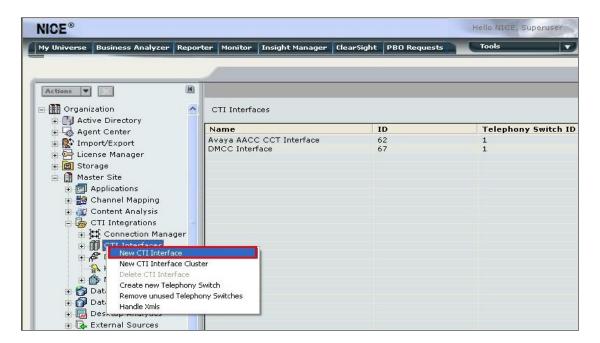
The previous section outlined the addition of a new CTI Connection which incorporates the setup of the following new configurations.

- CTI Interface
- Connection Manager
- Media Provider Controller

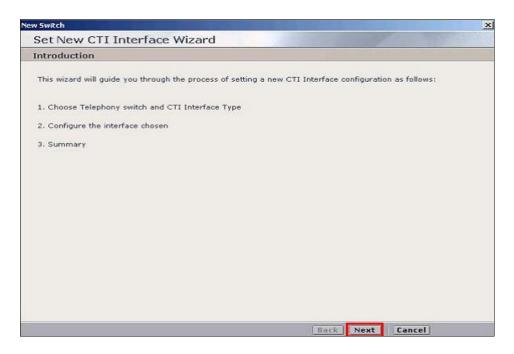
For Hybrid call recording or recording of non Contact Centre calls the DMCC must be configured to ensure that events from AES are being recorded. This means that a new CTI Interface, Connection Manager and Media Provider Controller must be setup.

#### 9.2.1. Configure a new CTI Interface

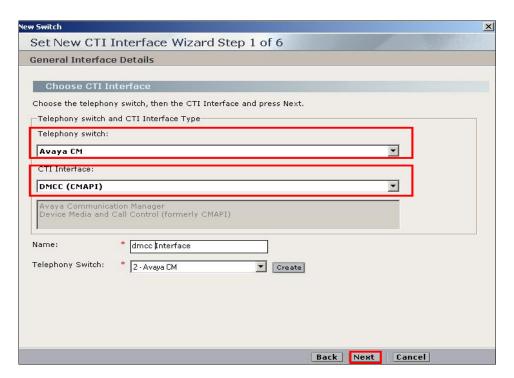
Navigate to Master Site → CTI Integrations → CTI Interfaces. Right click on CTI Interfaces and select New CTI Interface.



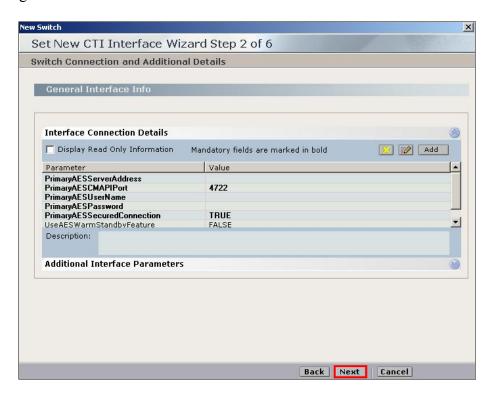
This brings up the window shown below. Click on Next to continue.



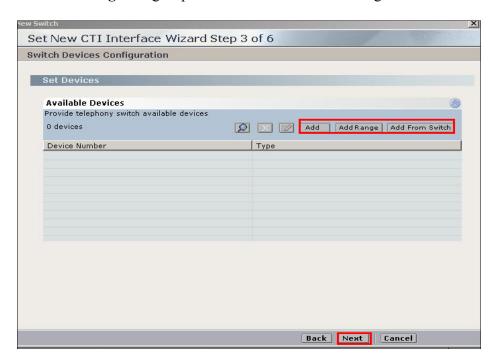
Ensure Avaya CM is selected for the Telephony switch and DMCC (CMAPI) for CTI Interface and click Next.



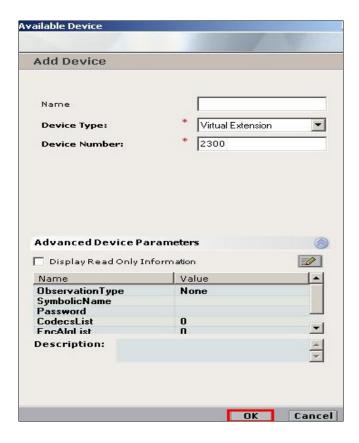
Enter the connection details for the AES (Username, Password, Domain and Address) to the AES as configured in Section 6.2.



Click on Add or Add Range if a group of extensions are to be configured.

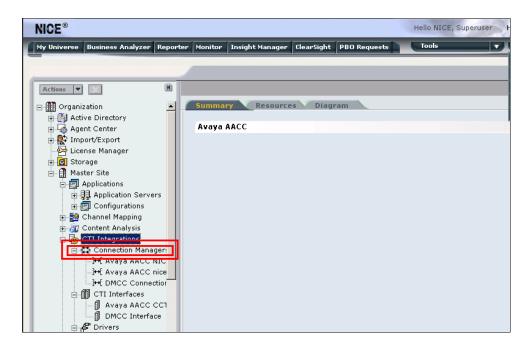


Enter a suitable Name, select Virtual Extension for Device Type and enter the extension number configured in Section 5.2 for the Device Number. Click OK when finished.

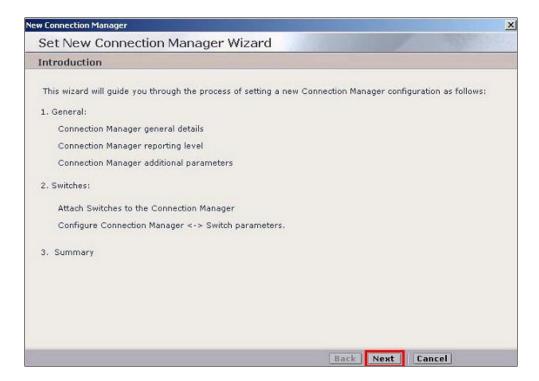


## 9.2.2. Configure new Connection Manager

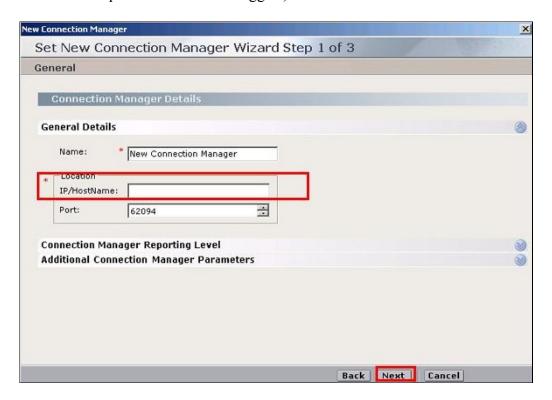
Navigate to **Master Site** → **CTI Integrations** → **Connection Manager**. Right click on **Connection Manager** and select New Connection (not shown).



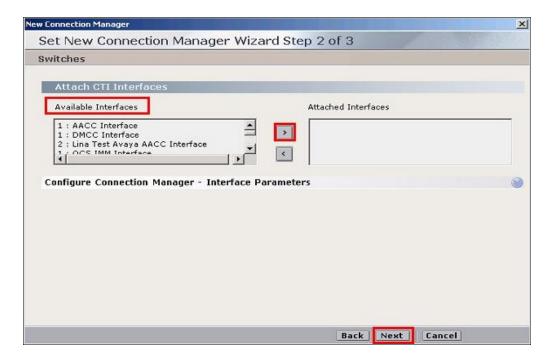
This brings up the window shown below. Click on Next to continue.



Enter a suitable name for the new connection and enter the IP address of the NICE logger server (see Section 9.3 for explanation of NICE Loggers) for IP/Hostname. Click on Next to continue.

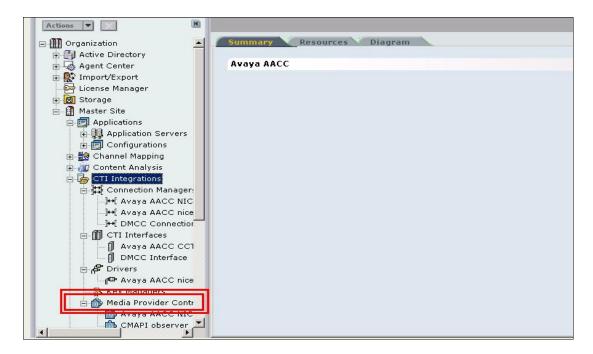


Select the DMCC interface as previously configured in **Section 9.2.1**, under **Available Interfaces**. Click on **Next** to continue.

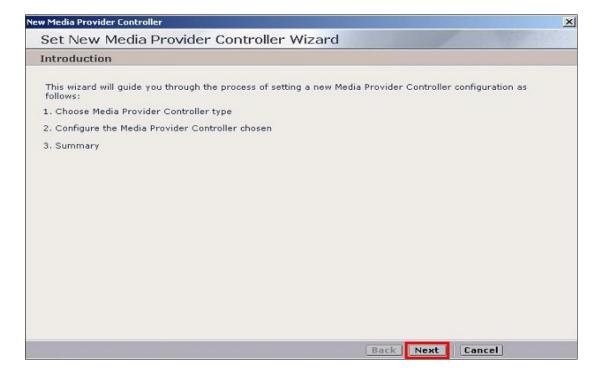


#### 9.2.3. Configure a new Media Provider Controller

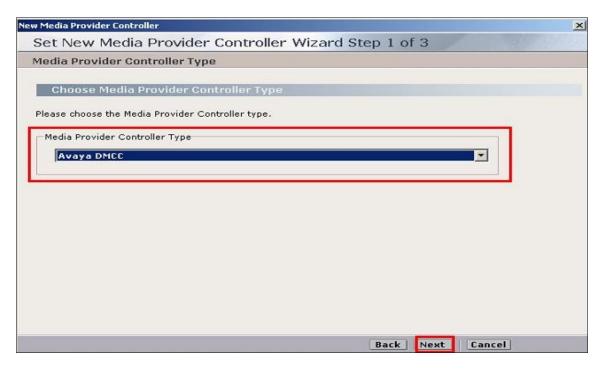
Navigate to Master Site → Media Provider Controllers. Right-click Media Provider Controllers and select New Media Provider Controller (not shown).



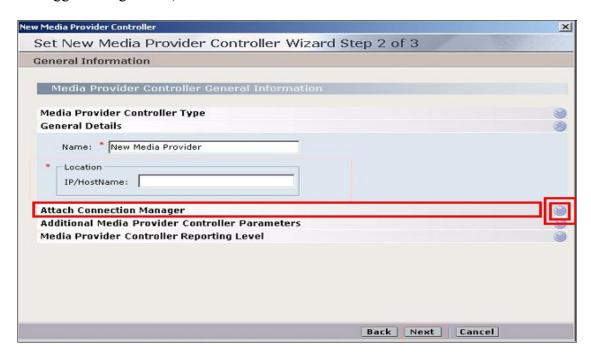
This brings up the window shown below. Click on **Next** to continue.



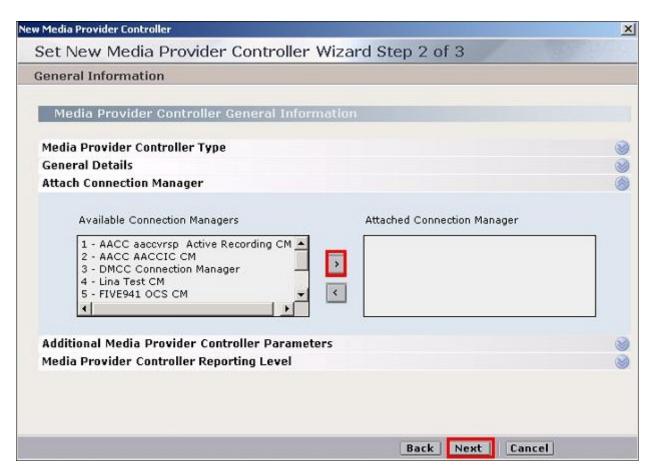
For the DMCC Media Provider Controller select Avaya DMCC for the Media Provider Controller Type. Click Next to continue.



Enter a suitable **Name** and enter the IP Address of the NICE Logger Server for **IP/Hostname**. The DMCC loggers IP address will be filled in for the DMCC Controller (see **Section 9.3** for the DMCC Logger configuration). Click **Next** to continue.



Expand the **Attach Connection Manager** highlighted on the previous page. This brings up the page shown below. In the **Available Connection Managers** list select the appropriate Connection Manager. For the DMCC controller, add the DMCC Connection Manager configured in **Section 9.2.2**. Click on **Next** to continue.

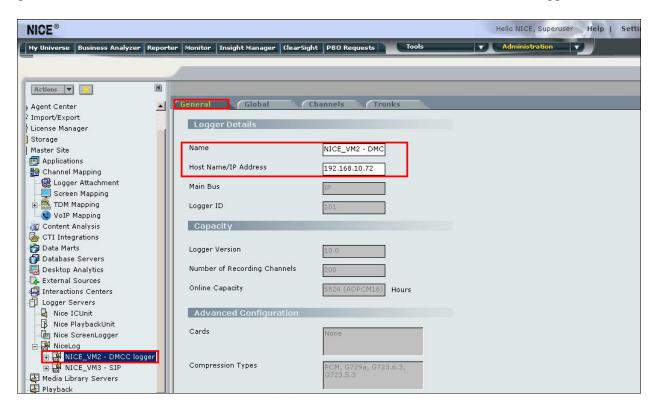


## 9.3. Configure NICE Loggers

NICE Loggers are responsible for the recording of voice calls, using a SIP based Logger for the recording of Contact Centre agent calls with events from CCT and using a DMCC based Logger for recording all other calls from the Communication Manager deskphones with events from AES. For this compliance testing these loggers are installed on separate servers. Configuration of these loggers is performed from the same management console as used in **Sections 9.1** and **9.2**.

**Note:** The types of Loggers are defined as a part of the install of NICE Interaction Management and therefore will not be covered in these Application Notes. For more information on this install please refer to **Section 12** of these Application Notes. However the configuration of the installed NICE Loggers is required and is explained below.

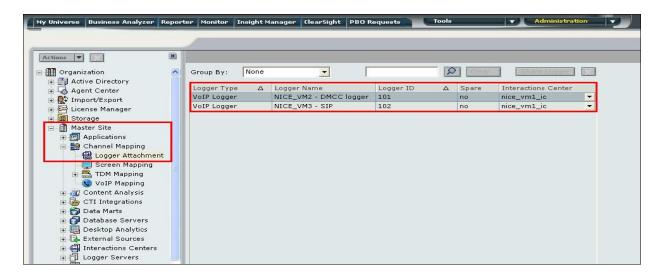
Navigate to **Logger Servers** → **Nice Log**. Select the Logger that is to be configured. Under the general tab enter a suitable **Name** and enter the **Host Name**/ **IP Address** of the Logger server.



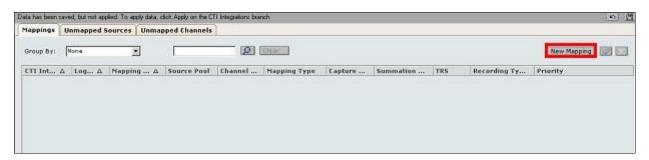
**Note:** Both the SIP and DMCC Loggers will need to be configured in order to successfully record both Contact Centre and Communication Manager calls.

# 9.4. Channel Mapping Configuration

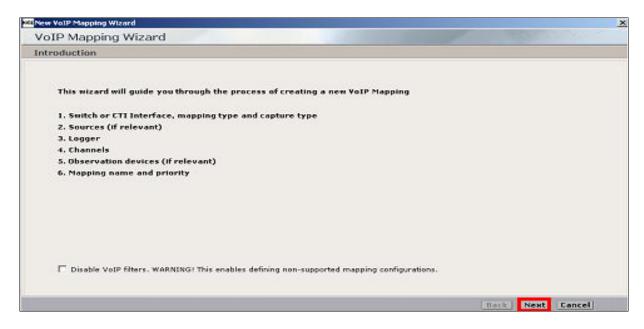
The Loggers configured above must be associated with an Interactions Center. Navigate to Master Site → Channel Mapping → Logger Attachment. This brings up the page shown below with a list of Voice, VoIP and NICE Screen Loggers. To associate an interactions centre with a Logger simply select the interaction centre required from the Interactions Center drop-down menu. Both the SIP and DMCC Loggers configured in Section 9.3 should be associated with an Interactions Center.



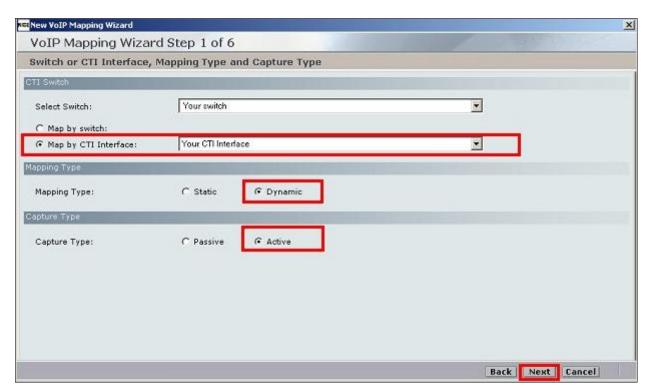
Navigate to **Master Site** → **Channel Mapping** → **VoIP Mapping**, which opens the window shown below. Click **New Mapping** highlighted to open the VoIP Mapping Wizard.



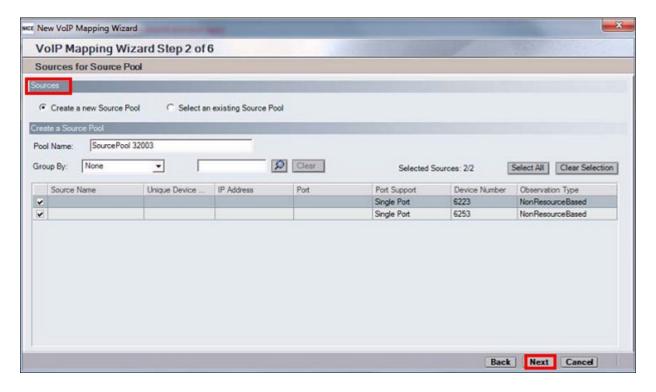
Click on **Next** to continue.



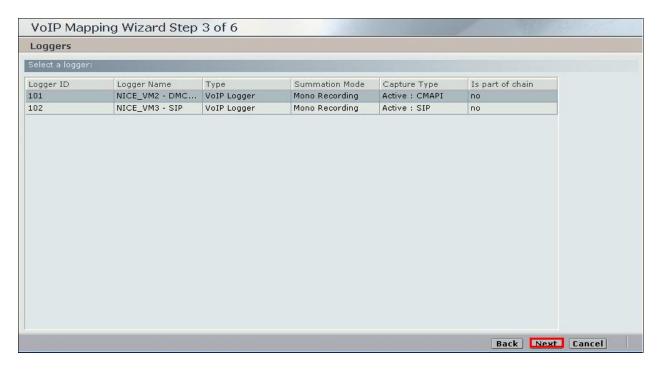
Ensure that **MAP by CTI Interface** is selected and that the relevant CTI Interface is chosen from the drop-down menu. For **Mapping Type**, select **Dynamic**. For **Capture Type**, select **Active**. Click **Next**. The Select Sources window appears.



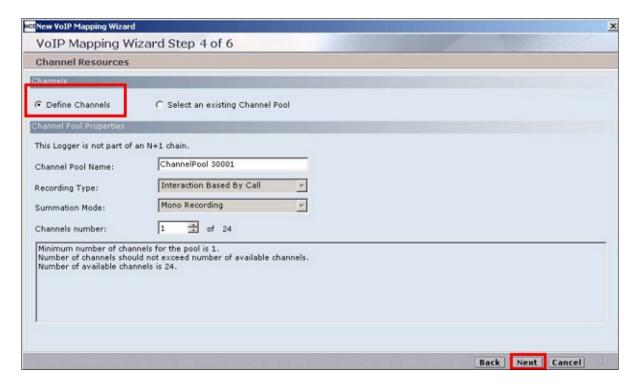
Select **Create a new Source Pool**. Enter a Suitable name for the Source Pool. By default, all devices are selected, however not all devices may be relevant for channel mapping.



Select the VoIP Logger for VoIP mapping. Click Next to continue.



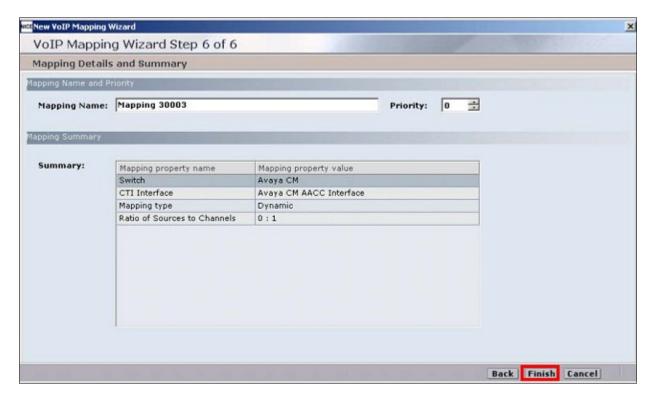
Select **Define Channels** and ensure that a suitable **Channel Pool Name** is chosen. In the **Number of Channels** field, enter the number of required channels for the pool, this determines the number of recordable devices in this Source Pool that can be simultaneously recorded. Click **Next** to continue.



In the **Mapping Name** field, enter a name for this channel mapping configuration. When necessary, set a **Priority**.

**Note:** Setting a **Priority** is a method of prioritizing recording resources for a specific mapping. This is relevant for site configurations in which one Source Pool is mapped to several Channel Pools or Loggers.

Click **Finish**. All of the devices in this defined Source Pool are mapped to VoIP Logger Recording channels.



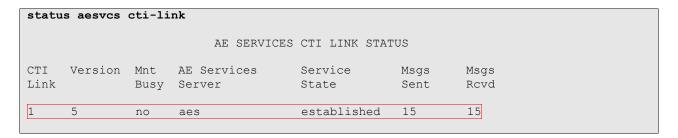
Note: This configuration is required for both the SIP and DMCC Channel Mapping.

# 10. Verification Steps

The following steps can be taken to ensure that connections between Communication Manager, AES, Contact Centre and NICE Interaction Management are configured correctly.

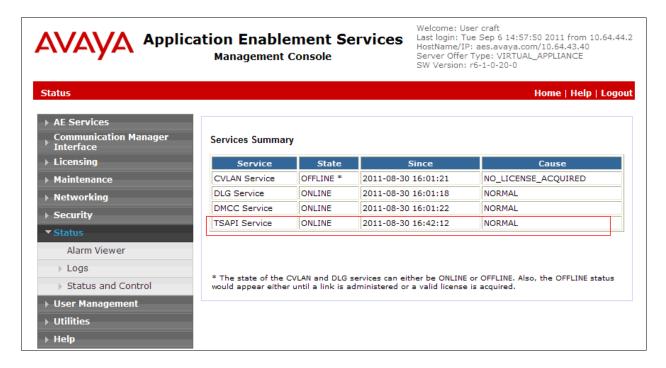
# 10.1. Verify Avaya Aura® Communication Manager CTI link

Verify the status of the administered CTI link by using the **status aesvcs cti-link** command. Verify the **Service State** is **established** for the CTI link number administered in **Section 5.1**, as shown below.



# 10.2. Verify Avaya Aura® Application Enablement Services CTI link

From the Application Enablement Services Management Console web pages, verify the state of the TSAPI Service is set to **ONLINE** by selecting **Status** from the left pane.



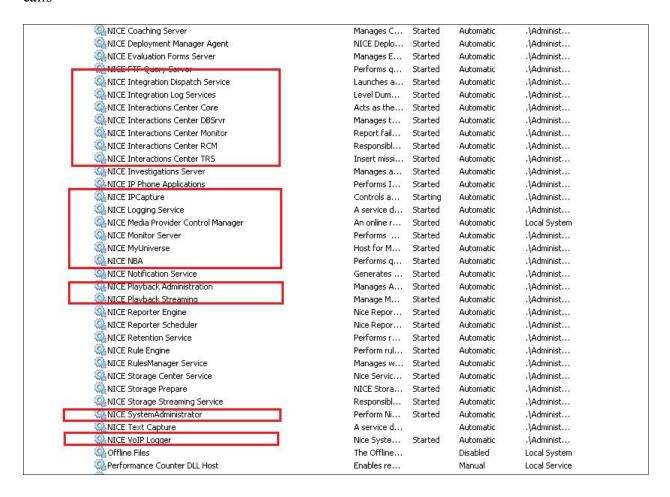
## 10.3. Verify that Web Services are running correctly

The following steps can be taken to ensure that Web Services are running correctly to allow NICE Interaction Management to receive call events to the API.

- 1. From any machine on the network, start Internet Explorer
- 2. Enter the following: http://<ccmaservername>/supportutil/testwebservices.asp
- 3. The web services that appear in green are running correctly

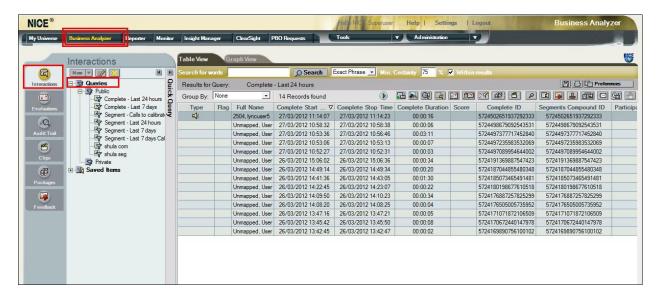
# 10.4. Verify NICE Interaction Management services are running

Go to Start → Services. Check the system services and make sure all NICE services are running. Highlighted below are a list of services that must be running to allow recording and playback of calls

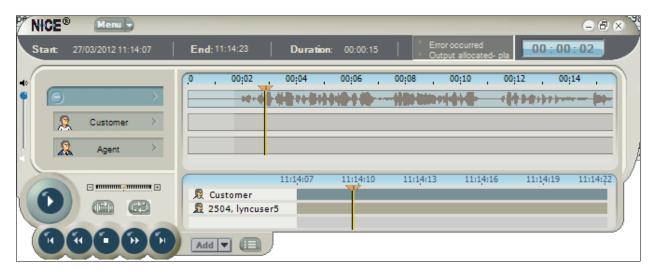


## 10.5. Verify calls are being recorded by NICE Interaction Management

Recordings are stored on the NICE Interaction Management server and can be replayed using **Business Analyzer** as shown below. Log in to NICE Interaction Management as shown in **Section 9.1**, click on the **Business Analyzer** tab. Select the Interactions tab on the left column and under **Queries** → **Public**, press on the query required.



Double click on any interaction in the table above to play it back. An example is shown below.



#### 11. Conclusion

These Application Notes describe the configuration steps required for NICE Interaction Management R4.1 to successfully interoperate with Avaya Aura® Contact Centre and Avaya Aura® Application Enablement Services in a Mission Critical High Availability Environment. All test cases were completed successfully. Please refer to **Section 2.2** for test results and High Availability failover observations.

#### 12. Additional References

This section references documentation relevant to these Application Notes. The Avaya product documentation is available at <a href="http://support.avaya.com">http://support.avaya.com</a> where the following documents can be obtained.

- [1] Administering Avaya Aura® Communication Manager, Document ID 03-300509
- [2] Avaya Aura® Communication Manager Feature Description and Implementation, Document ID 555-245-205
- [3] Avaya Aura® Application Enablement Services Administration and Maintenance Guide Release 6.1 Issue 2
- [4] Avaya Aura ® Contact Centre SIP Commissioning, Doc # NN44400-511, Issue 3.02 Release 6.2
- [5] Avaya Aura ® Contact Centre Planning and Engineering, Doc # NN44400-210, Issue 3.03 Release 6.2
- [6] Avaya Aura ® Contact Centre Installation, Doc #NN44400-311, Issue 3.02 Release 6.2

All information on the product installation and configuration of NICE Interaction Management can be found at <a href="http://www.nice.com">http://www.nice.com</a>

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