



## **Avaya Solution & Interoperability Test Lab**

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# **Application Notes for configuring NICE Engage Platform R6.3 to interoperate with Avaya Aura® Communication Manager R6.3 and Avaya Aura® Application Enablement Services R6.3 using DMCC Service Observation and Single Step Conference to record calls - Issue 1.0**

## **Abstract**

These Application Notes describe the configuration steps for the NICE Engage Platform to interoperate with the Avaya solution consisting of an Avaya Aura® Communication Manager R6.3, an Avaya Aura® Session Manager R6.3, and Avaya Aura® Application Enablement Services R6.3.

Readers should pay attention to Section 2, in particular the scope of testing as outlined in Section 2.1 as well as the observations noted in Section 2.2, to ensure that their own use cases are adequately covered by this scope and results.

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 configuration steps for the NICE Engage Platform R6.3 to interoperate with the Avaya solution consisting of an Avaya Aura® Communication Manager R6.3, an Avaya Aura® Session Manager R6.3, and Avaya Aura® Application Enablement Services R6.3. The NICE Engage Platform was setup to use Device Media Call Control Service Observation and Single Step Conference to record both internal and external calls on various Communication Manager endpoints, listed in **Section 4**.

Device Media Call Control (DMCC) works by allowing software vendors to create soft phones, in memory on a recording server, and use them to monitor and record other phones. This is purely a software solution and does not require telephony boards or any wiring beyond a typical network infrastructure.

The NICE Engage Platform is fully integrated into a LAN (Local Area Network), and includes easy-to-use Web based applications (i.e. Nice Application) that works with .NET framework and can be used to retrieve telephone conversations from a comprehensive long-term calls database. The NICE Engage Platform uses the Communication Manager feature “Service Observe” to observe a call on an extension; this way the call is recorded and can be played back at a later time. NICE can also conference into the call and record the call using this method. Both methods of call recording use virtual stations on Communication Manager in order to observe or conference into existing calls to record them.

The NICE Engage Platform contains tools for audio retrieval, centralized system security authorization, system control, and system status monitoring. Also included is a call parameters database (Nice Application Server) that tightly integrates via CTI link PABXs and ACD’s including optional advanced audio archive database management, search tools, a wide variety of Recording-on-Demand capabilities, and comprehensive long-term call database for immediate retrieval.

## 2. General Test Approach and Test Results

The interoperability compliance testing evaluated the ability of the NICE Engage Platform to carry out call recording in a variety of scenarios using DMCC Service Observation and Single Step Conference with AES and Communication Manager. A range of Avaya endpoints were used in the compliance testing all of which are listed in **Section 4**.

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 interoperability compliance test included both feature functionality and serviceability testing. The feature functionality testing focused on placing and recording calls in different call scenarios with good quality audio recordings and accurate call records. The tests included:

- **Inbound calls** – Test call recording for inbound calls to the Communication Manager from PSTN callers.
- **Outbound calls** – Test call recording for outbound calls from the Communication Manager to PSTN callers.
- **Hold/Transferred/Conference calls** – Test call recording for calls transferred to and in conference with PSTN callers.
- **EC500 Calls/Forwarded calls** - Test call recording for calls terminated on Avaya DECT handsets using EC500.
- **Call Park/Call Pickup** Test call recording for calls that are parked or picked up using Call Park and Call Pickup.
- **Calls to Elite Agents** – Test call recording for calls to Communication Manager agents logged into one-X® Agent.
- **Failover testing** - The behaviour of NICE Engage Platform under different simulated failure conditions on the Avaya platform will also be observed.

## 2.2. Test Results

Most functionality and serviceability test cases were completed successfully. The following observation was noted.

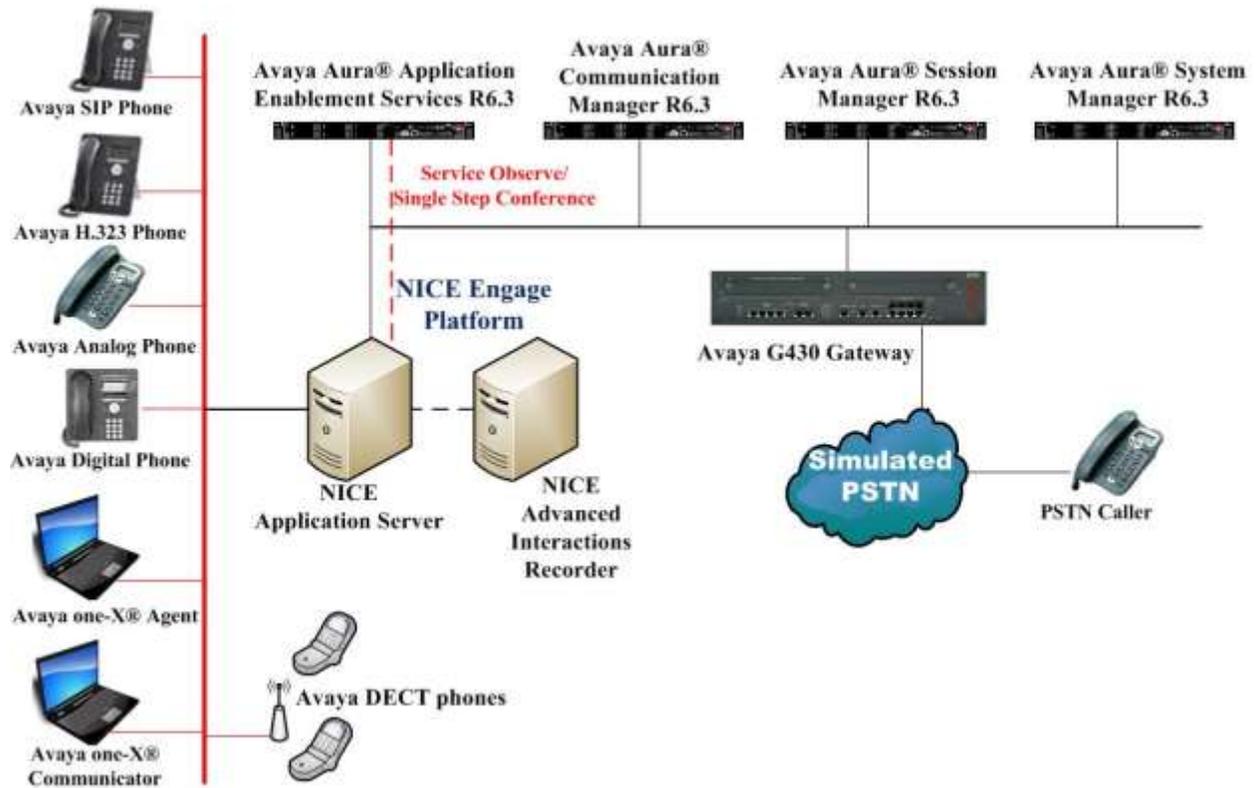
1. When transferring over a SIP trunk the recordings are presented for playback on the NICE GUI slightly differently than when transferring over a QSIG trunk and this is due to how the NICE recorder receives “updates” on the SIP trunk versus the QSIG. This is only an observation and the recording is just as valid.

## 2.3. Support

Technical support can be obtained for NICE Engage Platform from the website <http://www.nice.com/support-and-maintenance>

## 3. Reference Configuration

The configuration in **Figure 1** was used to compliance test NICE Engage Platform with the Avaya solution using DMCC Service Observe and Single Step Conference to record calls. The NICE Application Server is setup for DMCC Service Observe and Single Step Conference mode and connects to the AES.



**Figure 1: Connection of NICE Engage Platform R6.3 with Avaya Aura® Communication Manager R6.3, Avaya Aura® Session Manager R6.3 and Avaya Aura® Application Enablement Services R6.3**

## 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 Virtual Server	R6.3.10 [Build 6.3.0.8.5682-6.3.8.4514] [SW Update Rev 6.3.10.7.2656]
Avaya Aura® Session Manager running on Virtual Server	R6.3 (SP9) 6.3.9.0.639011
Avaya Aura® Communication Manager running on Virtual Server	R6.3 SP8 R016x.03.0.124.0 03.0.124.0-21588
Avaya Aura® Application Enablement Services running on Virtual Server	R6.3 Build No - 6.3.3.1.10-0
Avaya G430 Gateway	33.12.0 /1
Avaya 9608 H323 Deskphone	96xx H.323 Release 6.4014U
Avaya 9620 H323 Deskphone	R3.186A
Avaya 9641 SIP Deskphone	96x1-IPT-SIP-R6_4_1-081114
Avaya 9630 SIP Deskphone	R2.6.12.1
Avaya one-X® Communicator H.323	R6.2.4.07-FP4
Avaya one-X® Communicator SIP	R6.2.4.07-FP4
Avaya one-X® Agent	R 2.5.50022.0
Avaya 9408 Digital Deskphone Avaya Analog Deskphone	N/A
Avaya DECT Handsets	3725 DH4 (R3.3.11) 3720 DH3 (R3.3.11)
NICE Engage Platform - Application Server - Advanced Interactions Recorder - NDM Server	R6.3

## 5. Configure Avaya Aura® Communication Manager

The information provided in this section describes the configuration of Communication Manager relevant to this solution. For all other provisioning information such as initial installation and configuration, please refer to the product documentation in **Section 10**.

The configuration illustrated in this section was performed using Communication Manager System Administration Terminal (SAT).

### 5.1. Verify System Features

Use the **display system-parameters customer-options** command to verify that Communication Manager has permissions for features illustrated in these Application Notes. On **Page 3**, ensure that **Computer Telephony Adjunct Links?** is set to **y** as shown below.

```
display system-parameters customer-options                               Page 3 of 11
                                OPTIONAL FEATURES

Abbreviated Dialing Enhanced List? y          Audible Message Waiting? y
Access Security Gateway (ASG)? n              Authorization Codes? y
Analog Trunk Incoming Call ID? y              CAS Branch? n
A/D Grp/Sys List Dialing Start at 01? y      CAS Main? n
Answer Supervision by Call Classifier? y      Change COR by FAC? n
ARS? y                                         Computer Telephony Adjunct Links? y
ARS/AAR Partitioning? y                      Cvg Of Calls Redirected Off-net? y
ARS/AAR Dialing without FAC? y                DCS (Basic)? y
ASAI Link Core Capabilities? n                DCS Call Coverage? y
ASAI Link Plus Capabilities? n                DCS with Rerouting? y
Async. Transfer Mode (ATM) PNC? n
Async. Transfer Mode (ATM) Trunking? n        Digital Loss Plan Modification? y
ATM WAN Spare Processor? n                    DS1 MSP? y
ATMS? y                                       DS1 Echo Cancellation? y
Attendant Vectoring? y
```

### 5.2. Note procr IP Address for Avaya Aura® Application Enablement Services Connectivity

Display the procr IP address by using the command **display node-names ip** and note the IP address for the **procr** and AES (**aes63vmpg**).

```
display node-names ip                                                  Page 1 of 2
                                IP NODE NAMES

Name          IP Address
SM100         10.10.40.34
aes63vmpg    10.10.40.30
default       0.0.0.0
g430          10.10.40.15
procr        10.10.40.31
```

### 5.3. Configure Transport Link for Avaya Aura® Application Enablement Services Connectivity

To administer the transport link to AES use the **change ip-services** command. On **Page 1** add an entry with the following values:

- **Service Type:** Should be set to **AESVCS**.
- **Enabled:** Set to **y**.
- **Local Node:** Set to the node name assigned for the procr in **Section 5.2**
- **Local Port:** Retain the default value of **8765**.

```
change ip-services Page 1 of 4
```

IP SERVICES					
Service Type	Enabled	Local Node	Local Port	Remote Node	Remote Port
AESVCS	y	procr	8765		

Go to **Page 4** of the **ip-services** form and enter the following values:

- **AE Services Server:** Name obtained from the AES server, in this case **aes63vmpg**.
- **Password:** Enter a password to be administered on the AES server.
- **Enabled:** Set to **y**.

**Note:** The password entered for **Password** field must match the password on the AES server in **Section 6.2**. The **AE Services Server** should match the administered name for the AES server; this is created as part of the AES installation, and can be obtained from the AES server by typing **uname -n** at the Linux command prompt.

```
change ip-services Page 4 of 4
```

AE Services Administration				
Server ID	AE Services Server	Password	Enabled	Status
1:	aes63vmpg	*****	y	idle
2:				
3:				

### 5.4. Configure CTI Link for TSAPI Service

Add a CTI link using the **add cti-link n** command. Enter an available extension number in the **Extension** field. Enter **ADJ-IP** in the **Type** field, and a descriptive name in the **Name** field. Default values may be used in the remaining fields.

```
add cti-link 1 Page 1 of 3
```

CTI LINK	
CTI Link: 1	
Extension: 2002	
Type: ADJ-IP	
Name: aes63vmpg	COR: 1

## 5.5. Configure Communication Manager for Service Observation

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 x** to make a change to the Class or Restriction. This value needs to be enabled in order for Service Observe to work for call recording.

```

display cor 1                                     Page 1 of 23
                                     CLASS OF RESTRICTION
COR Number: 1
COR Description:

FRL: 0                                           APLT? y
Can Be Service Observed? y                   Calling Party Restriction: all-toll
Can Be A Service Observer? y                   Called Party Restriction: none
Time of Day Chart: 1                           Forced Entry of Account Codes? n
Priority Queuing? n                             Direct Agent Calling? y
Restriction Override: all                       Facility Access Trunk Test? n
Restricted Call List? n                        Can Change Coverage? n
Unrestricted Call List: 1
Access to MCT? y                               Fully Restricted Service? n
Group II Category For MFC: 7                   Hear VDN of Origin Annc.? n
Send ANI for MFE? n                            Add/Remove Agent Skills? n
MF ANI Prefix:                                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
  
```

Type **change feature-access-codes** to access the feature codes on Communication Manager. Scroll to **Page 5** in order to view or change the **Service Observing** access codes. Note the **Service Observing Listen Only Access Code** is **#43**; this will be required in **Section 7.1** during the setup of the NICE Engage Platform.

```

change feature-access-codes                       Page 5 of 10
                                     FEATURE ACCESS CODE (FAC)
                                     Call Center Features
AGENT WORK MODES
After Call Work Access Code: #36
Assist Access Code:
Auto-In Access Code: #38
Aux Work Access Code: #39
Login Access Code: #40
Logout Access Code: #41
Manual-in Access Code: #42
SERVICE OBSERVING
Service Observing Listen Only Access Code: #43
Service Observing Listen/Talk Access Code: #44
Service Observing No Talk Access Code:
Service Observing Next Call Listen Only Access Code:
Service Observing by Location Listen Only Access Code:
Service Observing by Location Listen/Talk Access Code:
AACC CONFERENCE MODES
Restrict First Consult Activation:              Deactivation:
Restrict Second Consult Activation:            Deactivation:
  
```

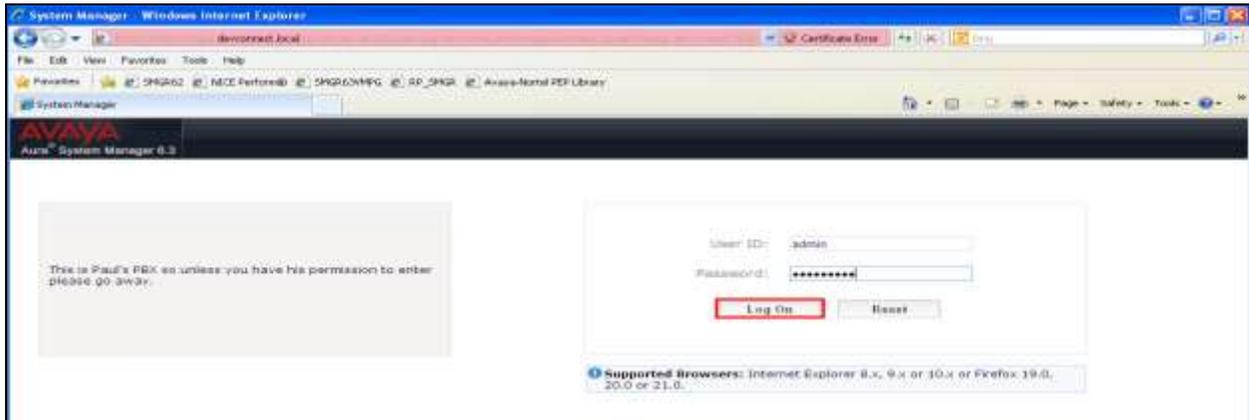
## 5.6. Configure H323 Stations for Service Observation

All endpoints that are to be monitored by NICE will need to have IP Softphone set to Y. IP Softphone must be enabled in order for DMCC Service Observe and Single Step Conference to work. Type **change station x** where x is the extension number of the station to be monitored also note this extension number for configuration required in **Section 7.1**. Note the **Security Code** and ensure that **IP SoftPhone** is set to y.

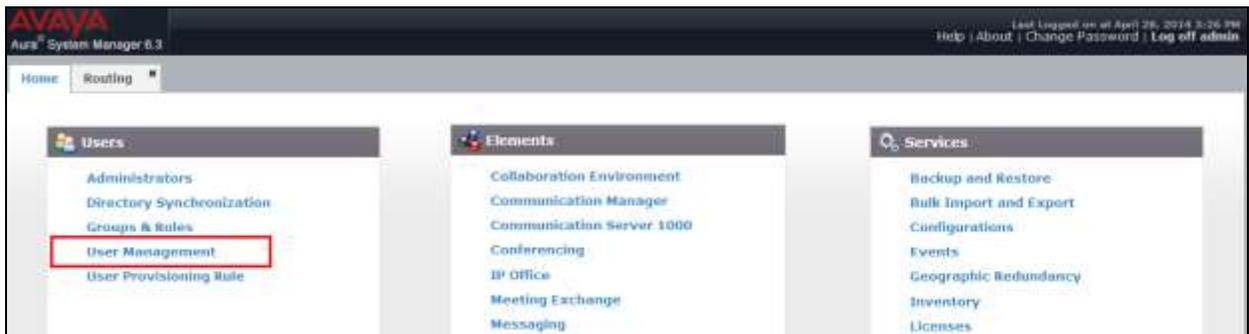
```
change station x                                     Page 1 of 6
                                                    STATION
Extension: x                                         Lock Messages? n      BCC: 0
  Type: 9630                                         Security Code: 1234  TN: 1
  Port: S00101                                       Coverage Path 1:      COR: 1
  Name: Recorder                                     Coverage Path 2:      COS: 1
                                                    Hunt-to Station:
STATION OPTIONS
  Loss Group: 19                                     Time of Day Lock Table:
  Speakerphone: 2-way                               Personalized Ringing Pattern: 1
  Display Language: english                         Message Lamp Ext: 1591
  Survivable GK Node Name:                          Mute Button Enabled? y
  Survivable COR: internal                           Media Complex Ext:
  Survivable Trunk Dest? y                           IP SoftPhone? y
                                                    IP Video Softphone? n
                                                    Short/Prefixed Registration Allowed: default
```

## 5.7. Configure SIP Stations for Service Observation

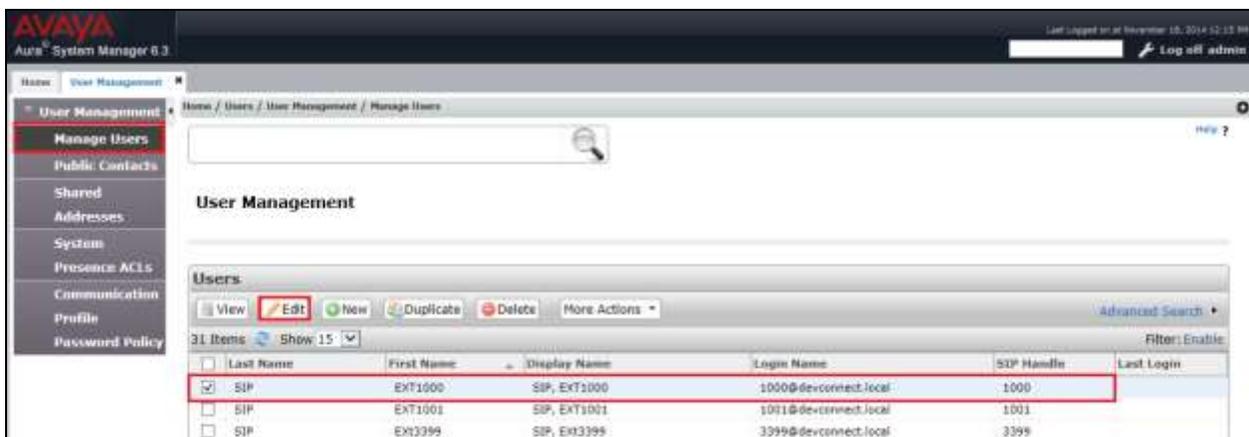
The configuration of SIP phones on Communication Manager must be carried out from System Manager. Access the System Manager using a Web Browser by entering **http://<FQDN>/SMGR**, where **<FQDN>** is the fully qualified domain name of System Manager or **http://<IP Address>/SMGR**. Log in using appropriate credentials.



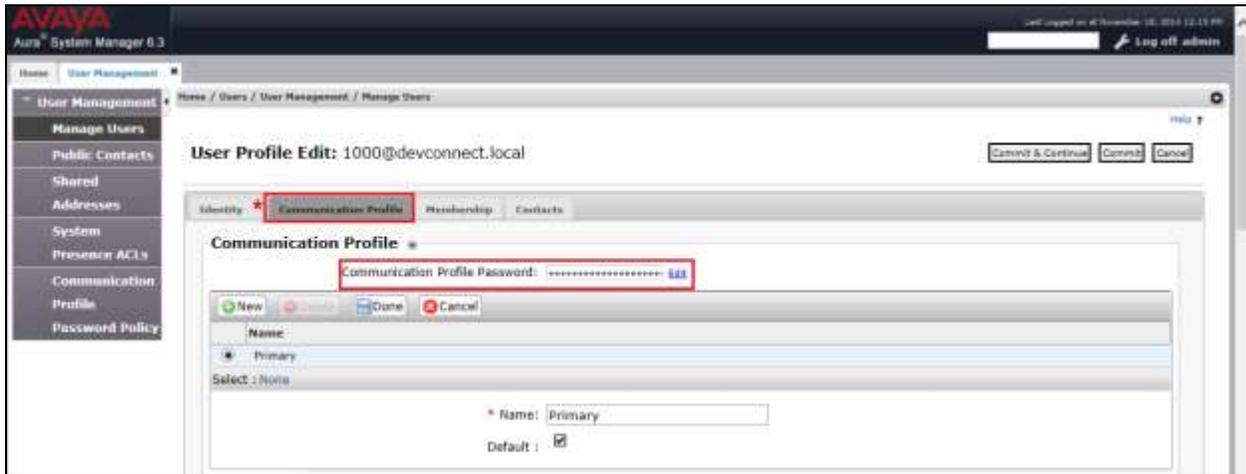
From the home page click on **User Management** highlighted below.



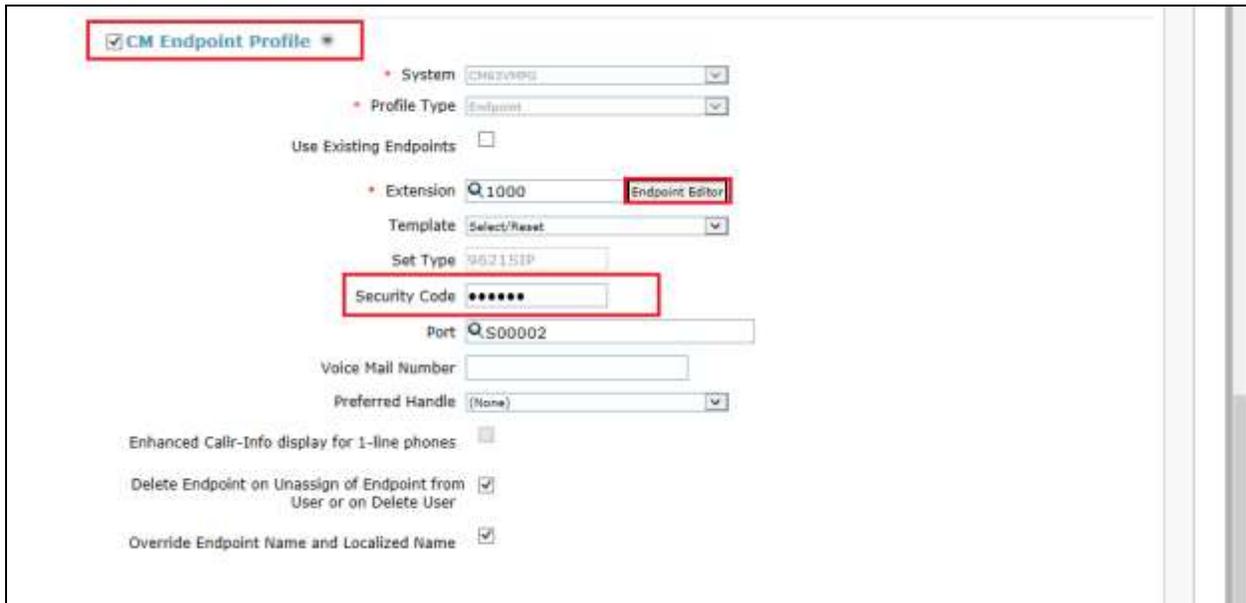
Click on **Manager Users** in the left window. Select the station to be edited and click on **Edit**.



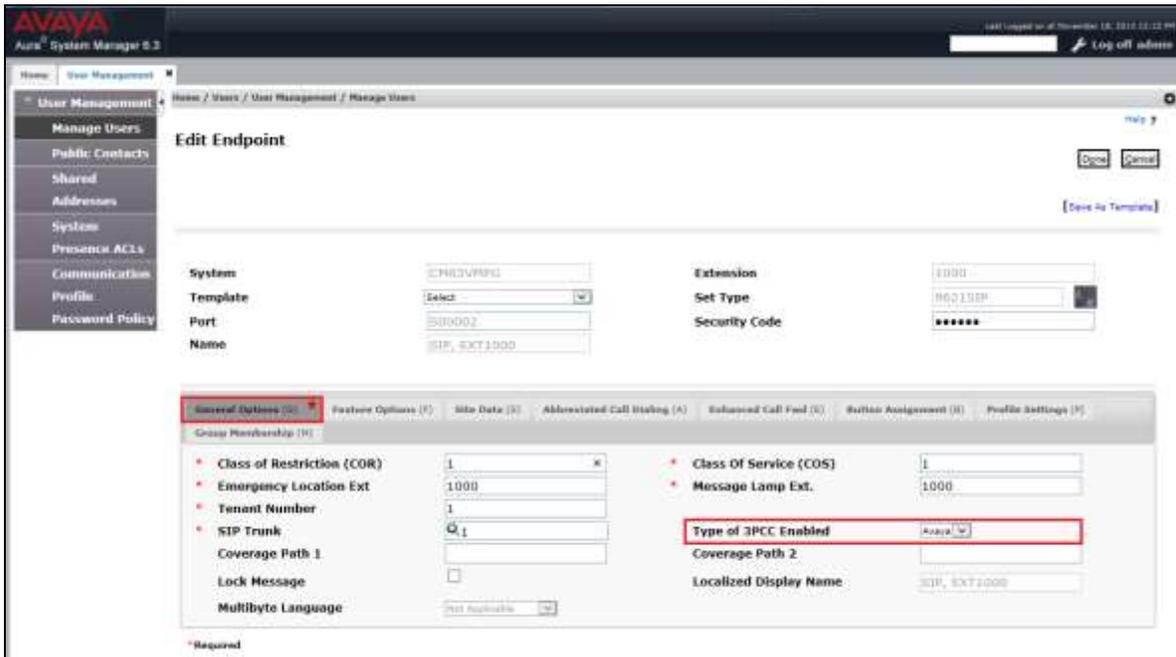
Click on the **Communication Profile** tab. Ensure that the **Communication Profile Password** is known and if not click on edit to change it.



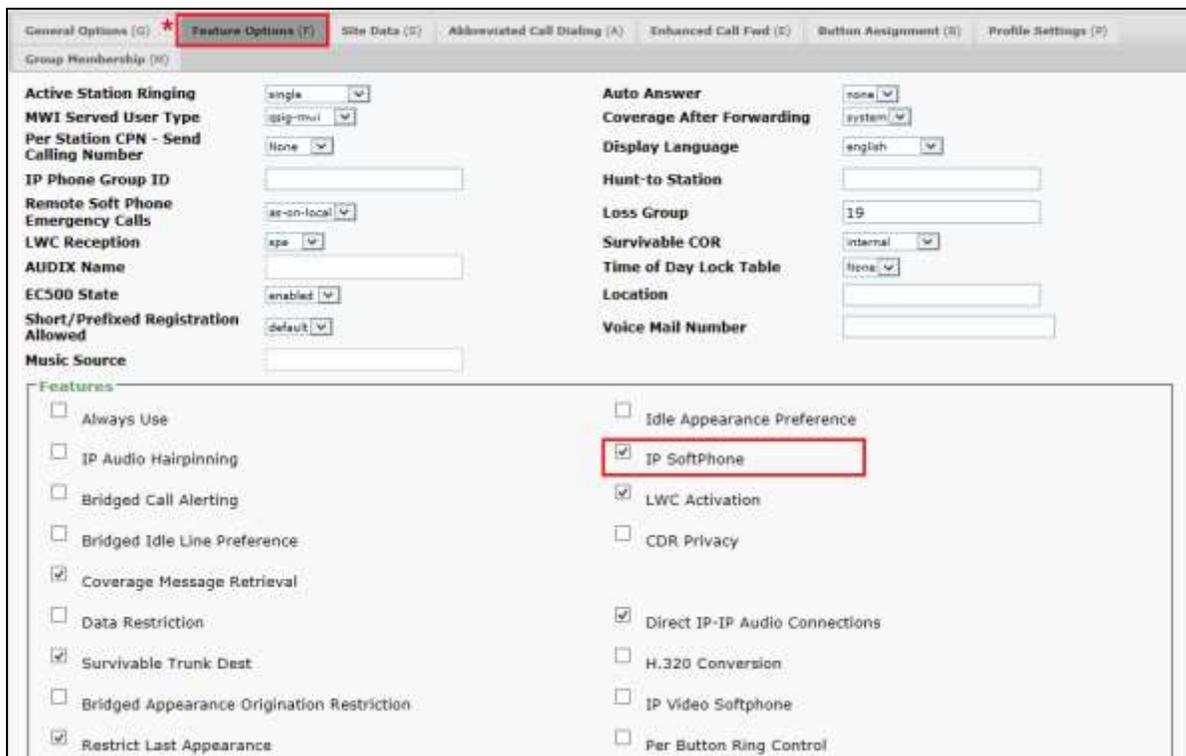
From the same page scroll down to **CM Endpoint Profile** and enter the **Security Code**, note this should be the same as the password above and will be required again in **Section 7.1** during the configuration of the NICE Engage Platform. Click on **Endpoint Editor** to make further changes.



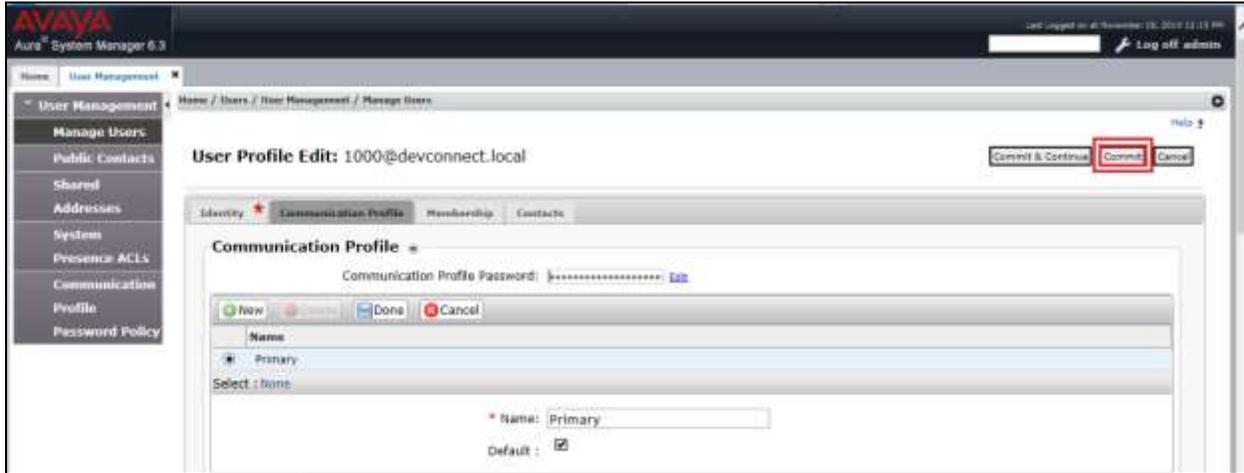
In the **General Options** tab ensure that **Type of 3PCC Enabled** is set to **Avaya** as is shown below. Also that Class of Restriction is set to that configured in **Section 5.5**.



Click on the **Feature Options** tab and ensure that **IP Softphone** is ticked as shown. Click on **Done** once this is set (not shown).



Click on **Commit** once this is done to save the changes.



## 5.8. Configure Virtual Stations for Single Step Conference and Service Observation

Add virtual stations to allow NICE Engage Platform record calls using Single Step Conference and 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 7.1**. Note the **Security Code** and ensure that **IP SoftPhone** is set to **y**. Note also the **COR** for the stations, this will be set to that configured in **Section 5.5**.

```

add station 28902                                     Page 1 of 6
                                                    STATION
Extension: 28902                                     Lock Messages? n                               BCC: 0
  Type: 4624                                         Security Code: 1234                             TN: 1
  Port: S00101                                       Coverage Path 1:                               COR: 1
  Name: Recorder                                       Coverage Path 2:                               COS: 1
                                                    Hunt-to Station:
STATION OPTIONS
  Loss Group: 19                                       Time of Day Lock Table:
  Speakerphone: 2-way                                   Personalized Ringing Pattern: 1
  Display Language: english                             Message Lamp Ext: 28902
  Survivable GK Node Name:                               Mute Button Enabled? y
  Survivable COR: internal                               Media Complex Ext:
  Survivable Trunk Dest? y                               IP SoftPhone? y
                                                    IP Video Softphone? n
                                                    Short/Prefixed Registration Allowed: default
  
```

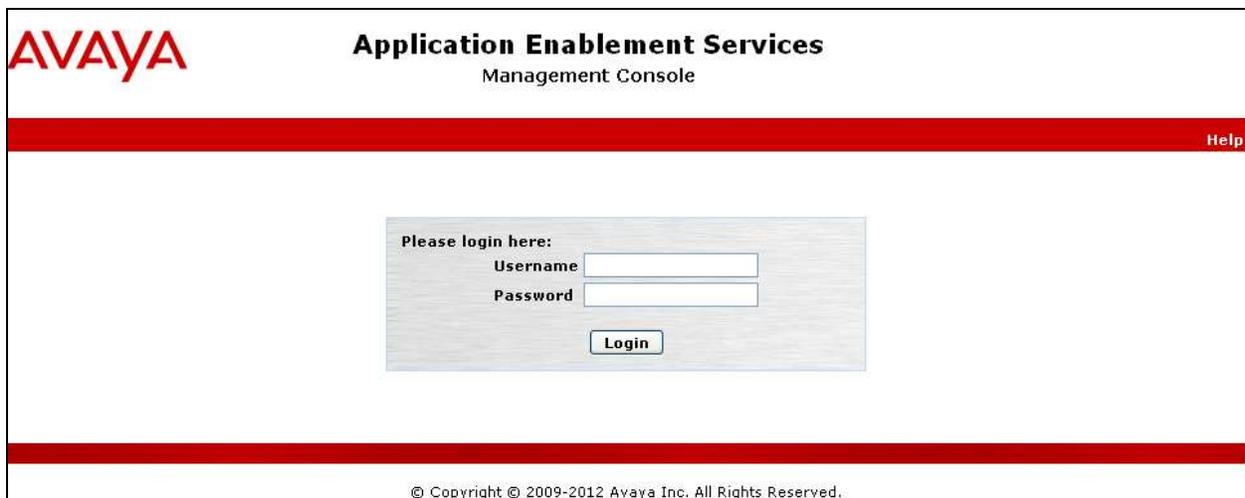
## 6. Configure Avaya Aura® Application Enablement Services

This section provides the procedures for configuring Application Enablement Services. The procedures fall into the following areas:

- Verify Licensing
- Create Switch Connection
- Administer TSAPI link
- Identify Tlinks
- Enable TSAPI Ports
- Create CTI User
- Set Up Security Database on AES
- Associate Devices with CTI User

### 6.1. Verify Licensing

To access the AES Management Console, enter **https://<ip-addr>** as the URL in an Internet browser, where <ip-addr> is the IP address of AES. At the login screen displayed, log in with the appropriate credentials and then select the **Login** button.



The screenshot shows the Avaya Application Enablement Services Management Console login page. The page features the Avaya logo in the top left corner, the title "Application Enablement Services Management Console" in the top center, and a "Help" link in the top right corner. A red horizontal bar spans the width of the page below the header. The main content area contains a login form with the text "Please login here:" followed by "Username" and "Password" labels, each with a corresponding text input field. Below the input fields is a "Login" button. A second red horizontal bar is located at the bottom of the page, above the copyright notice.

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The Application Enablement Services Management Console appears displaying the **Welcome to OAM** screen (not shown). Select **AE Services** and verify that the TSAPI Service is licensed by ensuring that **TSAPI Service** is in the list of **Services** and that the **License Mode** is showing **NORMAL MODE**. If not, contact an Avaya support representative to acquire the proper license for your solution.

AVAYA Application Enablement Services Management Console

Welcome! User: craft  
 Last login: Wed Dec 12 10:49:16 2012 from 193.168.10.208  
 Number of prior failed login attempts: 0  
 Hostname/IP: aes63/vmg.devconnect.local/10.10.40.10  
 Server Offer Type: SWONLY  
 SW Version: r5-2-0-1B-0  
 Server Date and Time: Thu Dec 20 11:51:09 UTC 2012

AE Services Home | Help | Logout

AE Services

IMPORTANT! AE Services must be restarted for administrative changes to fully take effect. Changes to the Security Database do not require a restart.

Service	Status	State	License Mode	Cause <sup>1</sup>
ASAI Link Manager	N/A	Running	N/A	N/A
CVLAN Service	ONLINE	Running	NORMAL MODE	N/A
DLG Service	ONLINE	Running	N/A	N/A
OHCC Service	ONLINE	Running	NORMAL MODE	N/A
TSAPI Service	ONLINE	Running	NORMAL MODE	N/A
Transport Layer Service	N/A	Running	N/A	N/A

For status of AOSM services, please use [Status and Control](#)

<sup>1</sup> - For more detail, please mouse over the Cause; you'll see the tooltip, or go to help page.

## 6.2. Create Switch Connection

From the AES Management Console navigate to **Communication Manager Interface** → **Switch Connections** to set up a switch connection. Enter a name for the Switch Connection to be added and click the **Add Connection** button.

AVAYA Application Enablement Services Management Console

Welcome! User: craft  
 Last login: Thu Nov 14 20:23:12 2013 from 10.10.40.140  
 Number of prior failed login attempts: 16  
 Hostname/IP: AES63YHFG  
 Server Offer Type: VIRTUAL\_APPLIANCE\_ON\_VMWARE  
 SW Version: 6.3.0.8.212-0  
 Server Date and Time: Tue Dec 3 15:33:26 UTC 2013

Communication Manager Interface | Switch Connections Home | Help | Logout

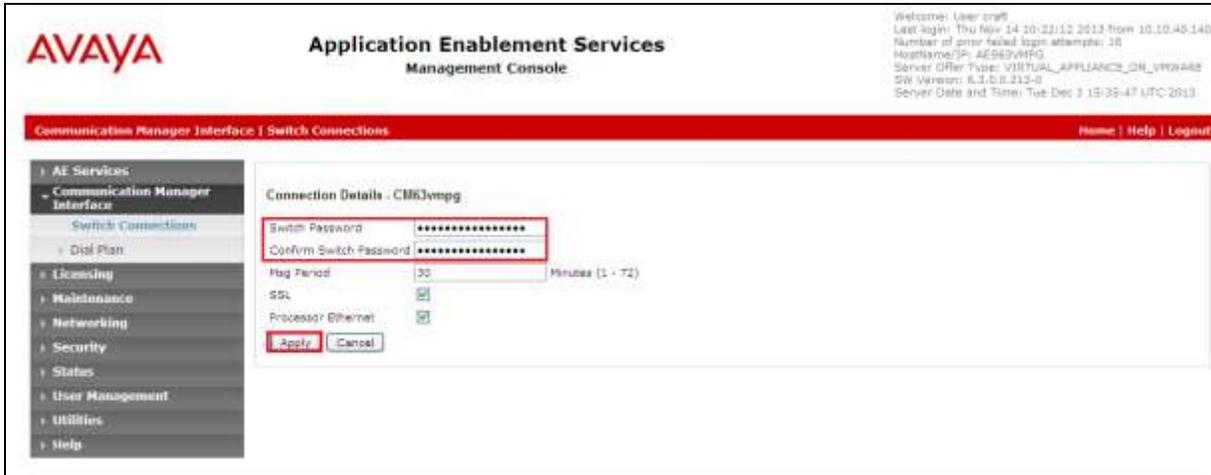
Switch Connections

COM2YHFG Add Connection

Connection Name	Processor Element	Reg Period	Number of Active Connections

Edit Connection Edit PE/CN IP Edit H.323 Gatekeeper Delete Connection Survivability Hierarchy

In the resulting screen enter the **Switch Password**; the Switch Password must be the same as that entered into Communication Manager AE Services Administration screen via the **change ip-services** command, described in **Section 5.3**. Default values may be accepted for the remaining fields. Click **Apply** to save changes.



From the **Switch Connections** screen, select the radio button for the recently added switch connection and select the **Edit PE/CLAN IPs** button (not shown, see screen at the bottom of the previous page). In the resulting screen, enter the IP address of the procr as shown in **Section 5.2** that will be used for the AES connection and select the **Add/Edit Name or IP** button.



### 6.3. Administer TSAPI link

From the Application Enablement Services Management Console, select **AE Services** → **TSAPI** → **TSAPI Links**. Select **Add Link** button as shown in the screen below.



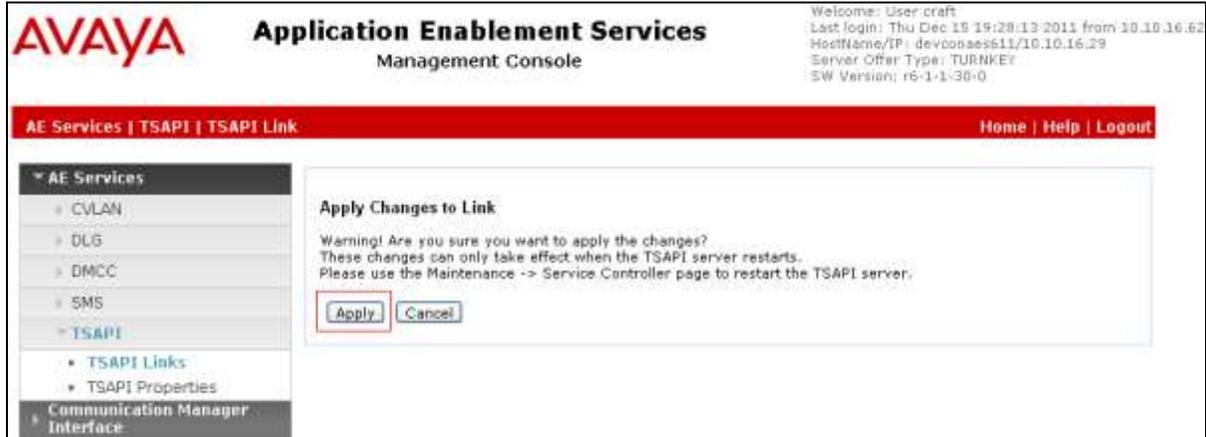
On the **Add TSAPI Links** screen (or the **Edit TSAPI Links** screen to edit a previously configured TSAPI Link as shown below), enter the following values:

- **Link:** Use the drop-down list to select an unused link number.
- **Switch Connection:** Choose the switch connection **CM63VMPPG**, which has already been configured in **Section 6.2** from the drop-down list.
- **Switch CTI Link Number:** Corresponding CTI link number configured in **Section 5.4** which is **1**.
- **ASAI Link Version:** This can be left at the default value of **5**.
- **Security:** This can be left at the default value of **both**.

Once completed, select **Apply Changes**.



Another screen appears for confirmation of the changes made. Choose **Apply**.



When the TSAPI Link is completed, it should resemble the screen below.



The TSAPI Service must be restarted to effect the changes made in this section. From the Management Console menu, navigate to **Maintenance** → **Service Controller**. On the Service Controller screen, tick the **TSAPI Service** and select **Restart Service**.



## 6.4. Identify Tlinks

Navigate to **Security** → **Security Database** → **Tlinks**. Verify the value of the **Tlink Name**. This will be needed to configure the Tlink Group in **Section 6.7.2**.

The screenshot displays the Avaya Application Enablement Services Management Console. The top left features the Avaya logo, and the top right shows the title "Application Enablement Services Management Console". A red navigation bar contains the text "Security | Security Database | Tlinks". On the left is a sidebar menu with categories: AE Services, Communication Manager Interface, Licensing, Maintenance, Networking, Security, Account Management, Audit, Certificate Management, Enterprise Directory, Host AA, PAM, Security Database, Control, CTI Users, Devices, Device Groups, and Tlinks (highlighted with a red box). The main content area is titled "Tlinks" and contains a "Tlink Name" section with two radio button options: "AVAYA#CM63VMPG#CSTA#AES63VMPG" (selected) and "AVAYA#CM63VMPG#CSTA-S#AES63VMPG". A "Delete Tlink" button is located below the options.

## 6.5. Enable TSAPI Ports

To ensure that TSAPI ports are enabled, navigate to **Networking** → **Ports**. Ensure that the TSAPI ports are set to **Enabled** as shown below. Ensure that the **DMCC Server Ports** are also **Enabled** and take note of the **Unencrypted Port 4721** which will be used later in **Section 7.1**.

The screenshot displays the Avaya Application Enablement Services Management Console. The interface includes a navigation menu on the left with categories like AE Services, Communication Manager, High Availability, Licensing, Maintenance, Networking, Security, Status, User Management, Utilities, and Help. The 'Networking' section is expanded to show 'Ports'. The main content area is titled 'Ports' and contains several sections:

- CVLAN Ports:** Includes 'Unencrypted TCP Port' (9999) and 'Encrypted TCP Port' (9998), both with 'Enabled' radio buttons selected.
- DLG Port:** Shows 'TCP Port' (5678).
- TSAPI Ports:** Includes 'TSAPI Service Port' (450) with an 'Enabled' radio button selected. Below it are 'Local TLINK Ports' (TCP Port Min: 1024, TCP Port Max: 1038) and 'Unencrypted TLINK Ports' (TCP Port Min: 1050, TCP Port Max: 1065).
- DMCC Server Ports:** Includes 'Unencrypted Port' (4721), 'Encrypted Port' (4722), and 'TR/87 Port' (4723), all with 'Enabled' radio buttons selected.

Red boxes highlight the 'Ports' menu item, the 'TSAPI Service Port' and 'Unencrypted Port' fields, and their respective 'Enabled' radio buttons.

## 6.6. Create CTI User

A User ID and password needs to be configured for the NICE Engage Platform to communicate with the Application Enablement Services server. Navigate to the **User Management** → **User Admin** screen then choose the **Add User** option.

The screenshot displays the Avaya Application Enablement Services Management Console. The top left features the AVAYA logo. The main title is "Application Enablement Services Management Console". A red navigation bar at the top indicates the current path: "User Management | User Admin".

On the left is a sidebar menu with the following items:

- ▶ AE Services
- ▶ Communication Manager Interface
- ▶ Licensing
- ▶ Maintenance
- ▶ Networking
- ▶ Security
- ▶ Status
- ▼ User Management
  - ▶ Service Admin
  - ▼ User Admin
    - Add User
    - Change User Password
    - List All Users
    - Modify Default Users
    - Search Users
- ▶ Utilities
- ▶ Help

### User Admin

User Admin provides you with the following options for managing AE Services users:

- Add User
- Change User Password
- List All Users
- Modify Default User
- Search Users

In the **Add User** screen shown below, enter the following values:

- **User Id** - This will be used by the NICE Engage Platform setup in **Section 7.1**.
- **Common Name** and **Surname** - Descriptive names need to be entered.
- **User Password** and **Confirm Password** - This will be used with NICE Engage Platform setup in **Section 7.1**.
- **CT User** - Select **Yes** from the drop-down menu.

Complete the process by choosing **Apply** at the bottom of the screen (not shown).

AVAYA Application Enablement Services Management Console

Welcome: user user  
Last login: Thu Nov 27 15:38:43 2014 from 10.10.60.50  
Number of prior failed login attempts: 0  
HostName/IP: AESSEVMFG10.10.40.30  
Server Offer Type: VIRTUAL\_APPLIANCE\_ON\_VMWARE  
SW Version: 8.3.1.1.10-0  
Server Date and Time: Mon Dec 01 16:03:36 GMT 2014  
HA Status: Not Configured

User Management | User Admin | List All Users Home | Help | Logout

AE Services  
Communication Manager Interface  
High Availability  
Licensing  
Maintenance  
Networking  
Security  
Status  
\* User Management  
Service Admin  
\* User Admin  
Add User  
Change User Password  
List All Users  
Modify Default Users  
Search Users  
Utilities  
Help

Edit User

\* User Id nice  
\* Common Name nice  
\* Surname nice  
User Password  
Confirm Password  
Admin Note  
Avaya Role None  
Business Category  
Car License  
CM Home  
CS Home  
CT User Yes  
Department Number  
Display Name  
Employee Number  
Employee Type

Scroll down and click on **Apply Changes**.

CM Home  
CS Home  
CT User Yes  
Department Number  
Display Name  
Employee Number  
Employee Type  
Enterprise Handle  
Given Name  
Home Phone  
Home Postal Address  
Initials  
Labeled URI  
Mail  
MN Home  
Mobile  
Organization  
Pager  
Preferred Language English  
Room Number  
Telephone Number

Apply Changes Cancel Changes

## 6.7. Associate Devices with CTI User

Navigate to **Security** → **Security Database** → **CTI Users** → **List All Users** and click on **Edit Users**.

The screenshot shows the Avaya Application Enablement Services Management Console. The breadcrumb trail is Security > Security Database > CTI Users > List All Users. A table lists CTI users with columns for User ID, Common Name, Worktop Name, and Device ID. The user 'nice' is selected, and the 'Edit' button is highlighted.

User ID	Common Name	Worktop Name	Device ID
<input type="radio"/> asc	asc	NONE	NONE
<input type="radio"/> cube	cube	NONE	NONE
<input type="radio"/> emc	emc	NONE	NONE
<input type="radio"/> jacada	jacada	NONE	NONE
<input checked="" type="radio"/> nice	nice	NONE	NONE
<input type="radio"/> presence	presence	NONE	NONE

In the main window ensure that **Unrestricted Access** is ticked. Once this is done click on **Apply Changes**.

The screenshot shows the 'Edit CTI User' page for the user 'nice'. The 'Unrestricted Access' checkbox is checked and highlighted. The 'Apply Changes' button is also highlighted.

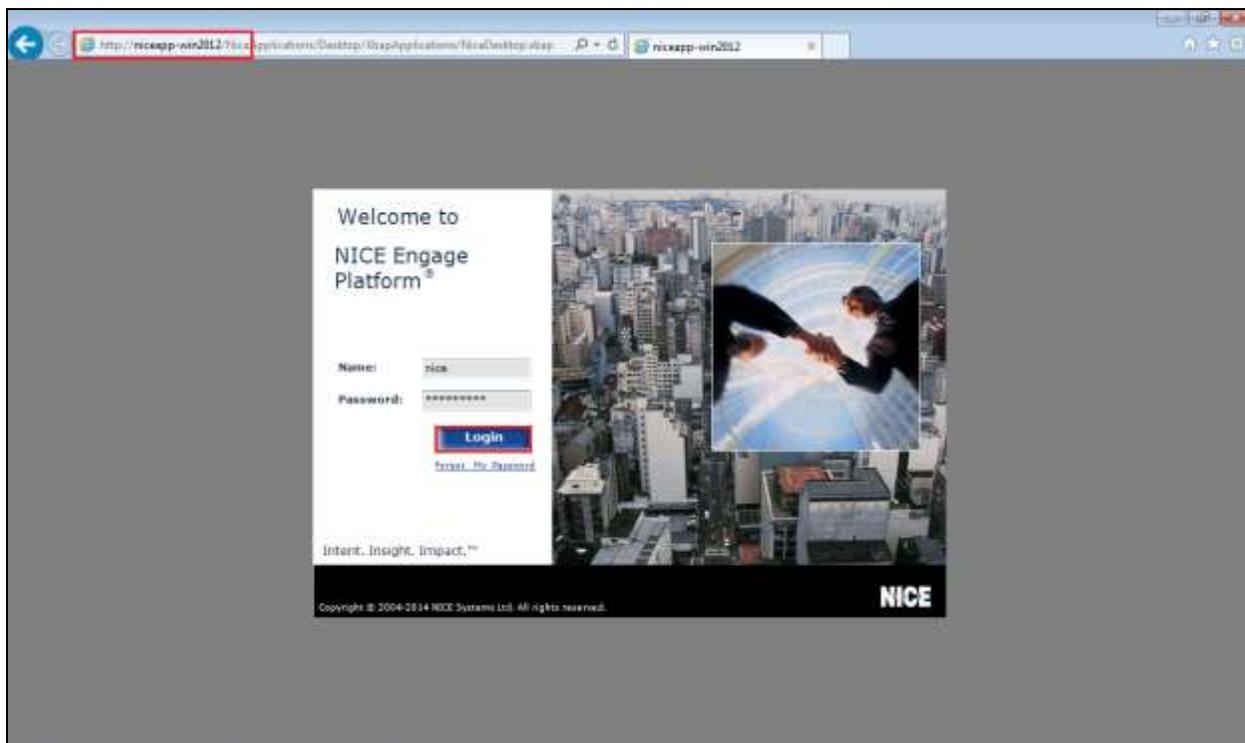
Field	Value
User ID	nice
Common Name	nice
Worktop Name	NONE
Unrestricted Access	<input checked="" type="checkbox"/>
Call and Device Control: Call Origination/Termination and Device Status	None
Call and Device Monitoring: Device Monitoring	None
Call and Device Monitoring: Calls On A Device Monitoring	None
Call and Device Monitoring: Call Monitoring	<input type="checkbox"/>
Routing Control: Allow Routing on Listed Devices	None

## 7. Configure NICE Engage Platform

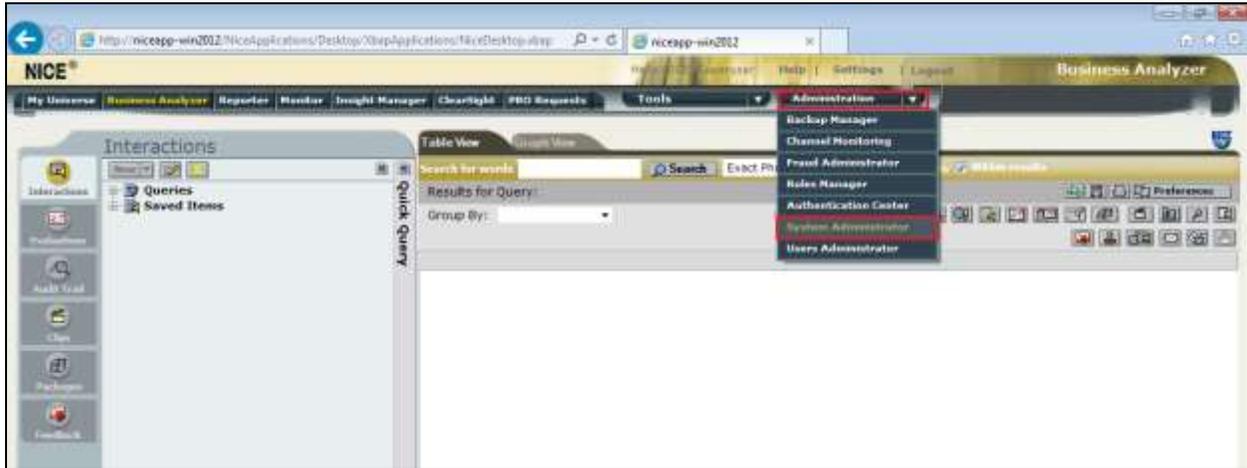
The installation of NICE Engage Platform is usually carried out by an engineer from NICE and is outside the scope of these Application Notes. For information on the installation of the NICE Engage Platform contact NICE as per the information provided in Section 2.3.

The following sections will outline the process involved in connecting the NICE Engage Platform to the Avaya Solution. All configuration of the NICE Engage Platform for connection with the AES is performed using a web browser connecting to the NICE Engage Application Server. Open a web browser as shown navigate to

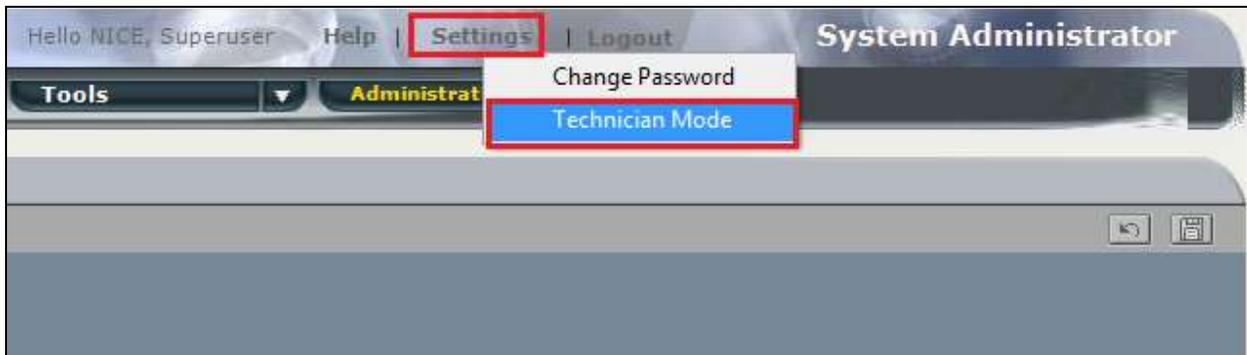
**http://<NICEEngageApplicationServerIP>/Nice** as shown below and enter the proper credentials and click on **Login**.



Once logged in expand the **Administration** dropdown menu and click on **System Administrator** as highlighted.

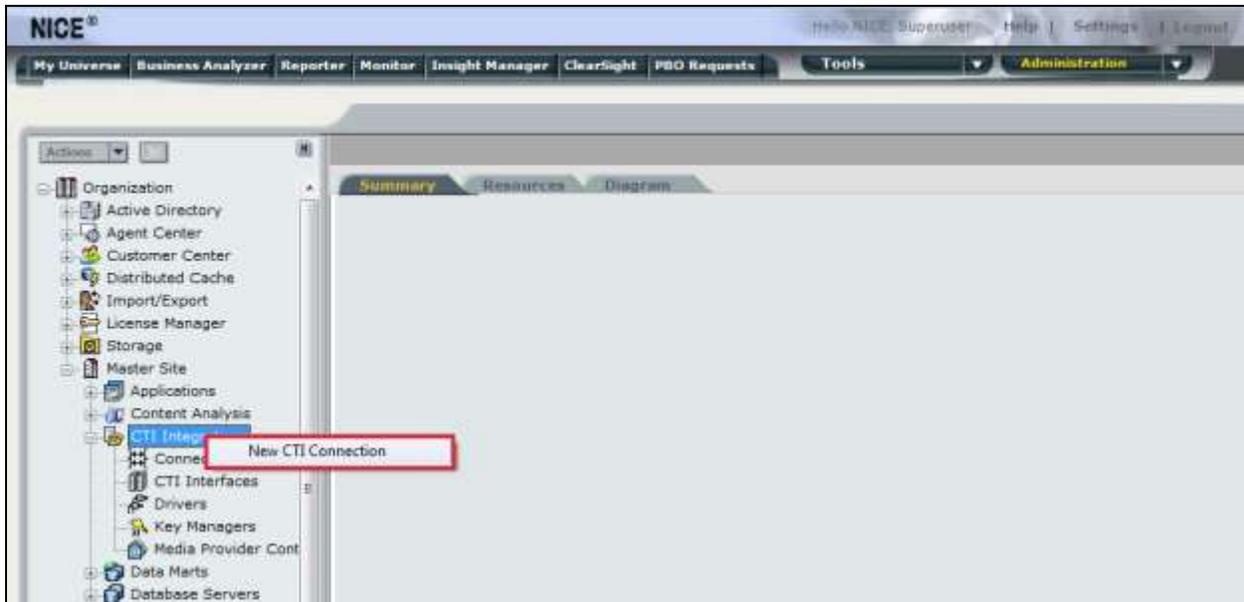


Before any changes can be made, switch to **Technician Mode** by clicking into **Settings** at the top of the screen as shown below.

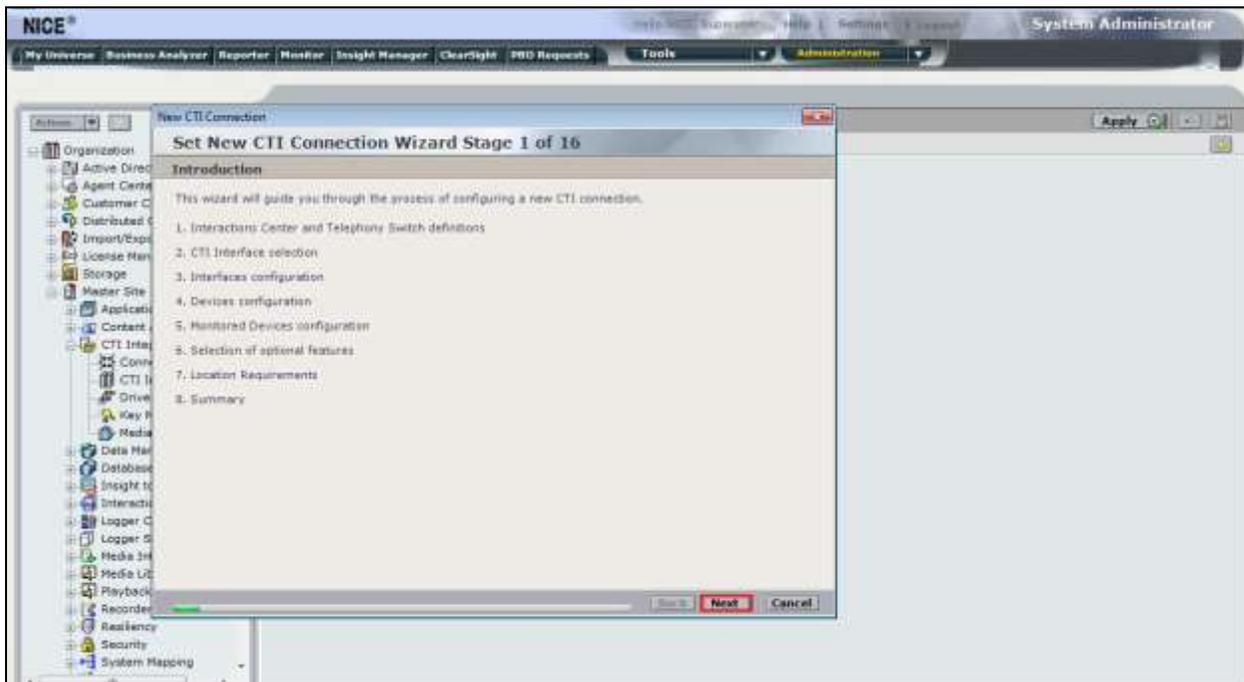


## 7.1. New CTI Connection

Navigate to **Master Site** → **CTI Integration** in the left window then right-click on CTI Integration and select **New CTI Connection** as shown below.



The **New CTI Connection Wizard** is opened and this will go through the 16 steps required to setup the connection to the AES for DMCC Service Observe and Single Step Conference type of call recording. Click on **Next** to continue.

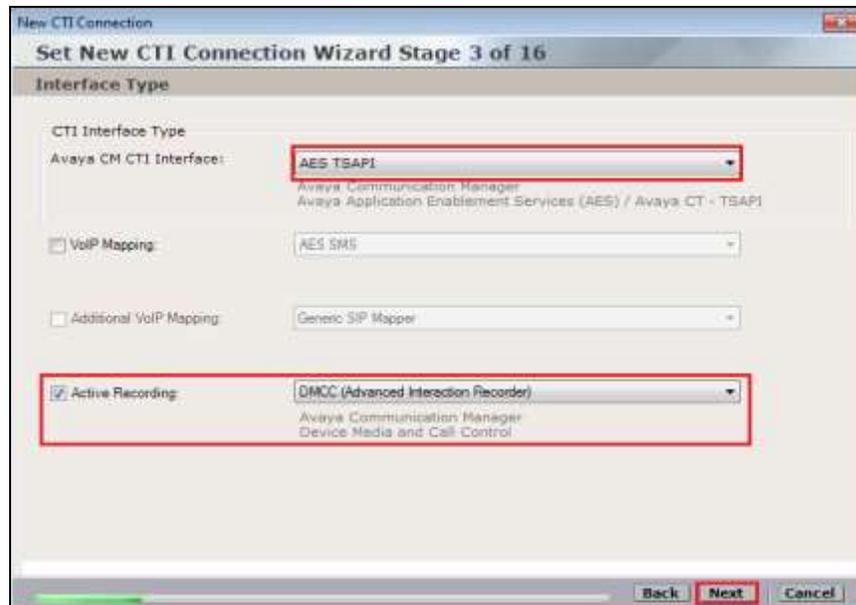


The value for Regular Interactions Center is a value that was already created during the installation of the NICE Engage platform. This value is therefore pre-chosen for the CTI connection being created below.

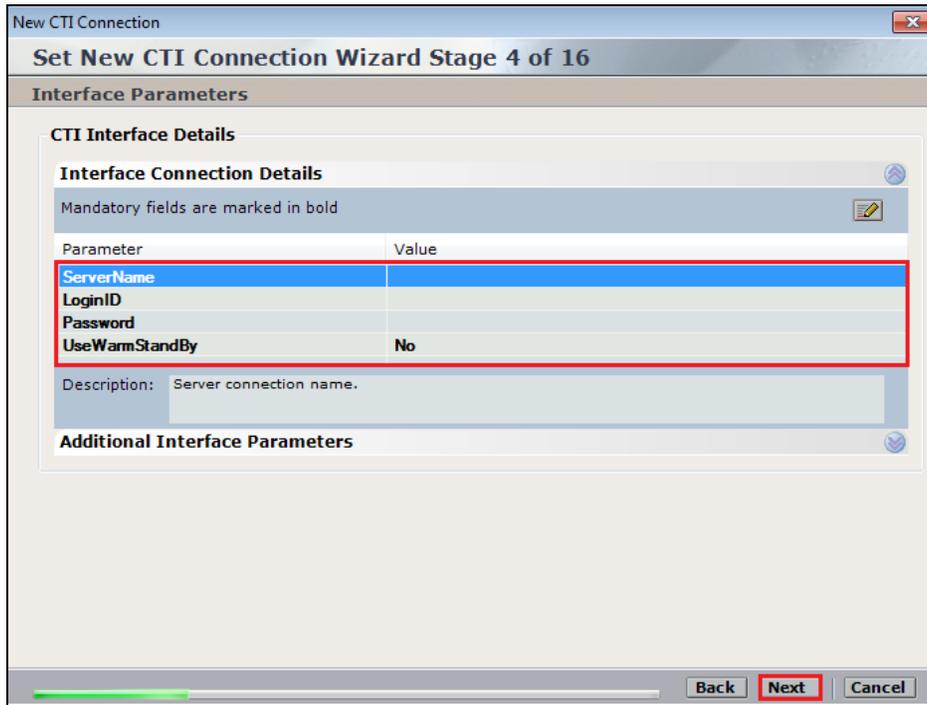
The **Telephony Switch** must be selected and this will be **Avaya CM**. Enter a suitable name for this **Switch Name**. Click on **Next** to continue.



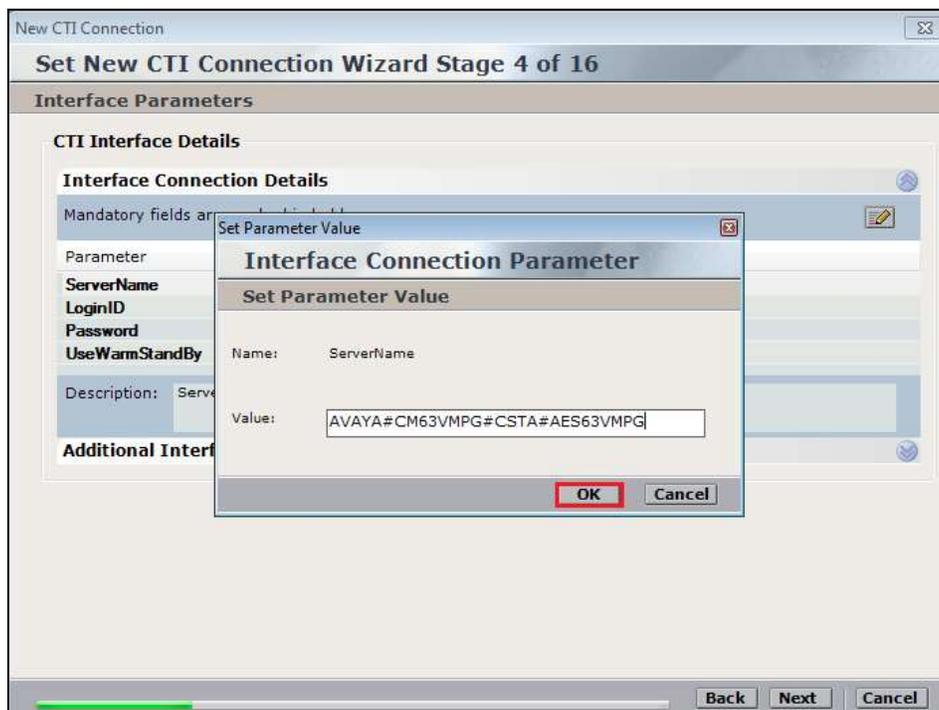
Select **AES TSAPI** for the **Avaya CM CTI Interface**, ensure that **Active Recording** is ticked and select the **DMCC (Advanced integration Recorder)** from the dropdown menu. Click on **Next** to continue.



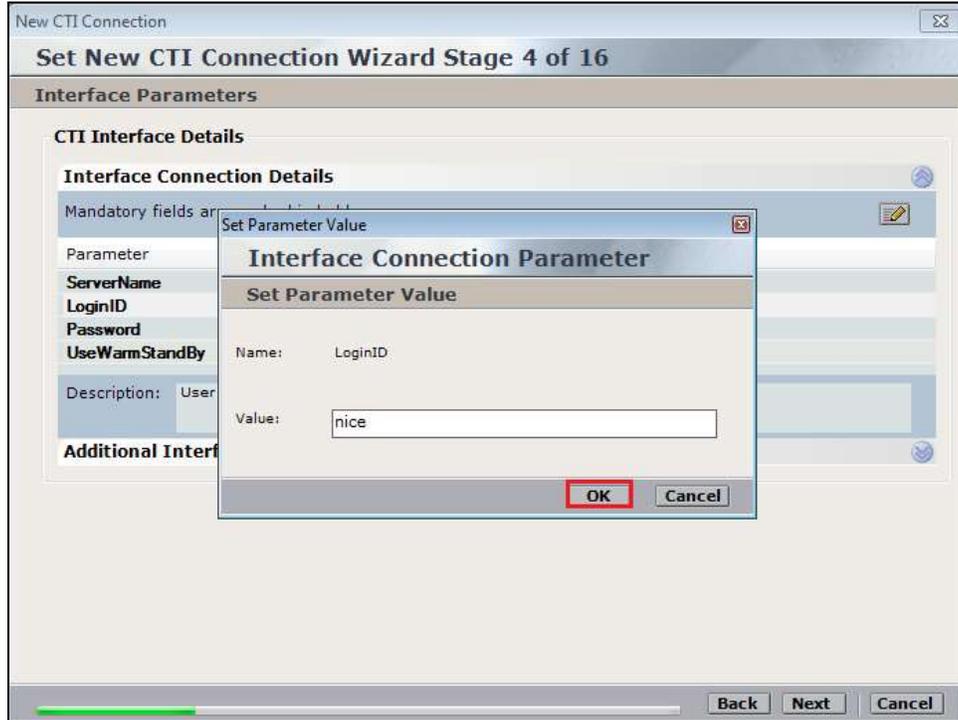
Each of the values below must be filled in. Double-click on each **Parameter** to enter a value for that parameter.



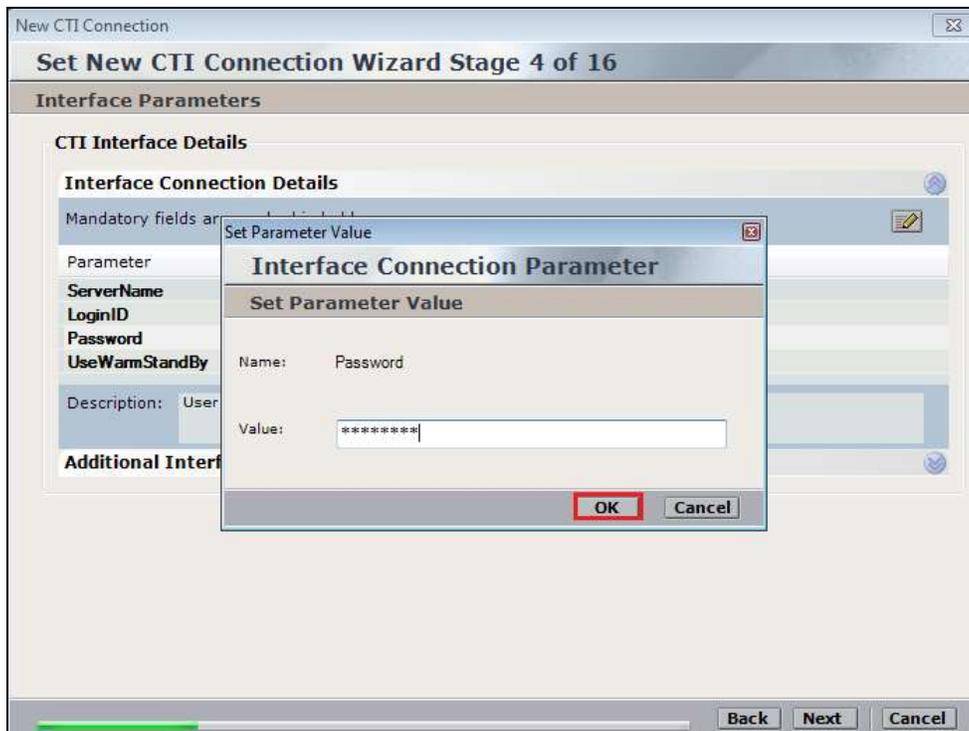
Double-click on **ServerName** and enter the TSAPI link **Value** from **Section 6.4**.



Double-click on LoginID and enter the username that was created in **Section 6.6**. Click on **OK**.



Double-click on password and enter the value for the password that was created in **Section 6.6**.



Click on **Next** once these values are all filled in.

The screenshot shows the 'Set New CTI Connection Wizard Stage 4 of 16' window. The 'Interface Parameters' section is active, displaying 'CTI Interface Details'. A table lists parameters and their values, with the 'UseWarmStandBy' row highlighted in blue. The 'Next' button is highlighted in red.

Parameter	Value
<b>ServerName</b>	AVAYA#CM63VMPPG#CSTA#AES63VMPPG
<b>LoginID</b>	nice
<b>Password</b>	*****
<b>UseWarmStandBy</b>	No

Description: Is warm standby supported?

Additional Interface Parameters

Back Next Cancel

The values below must be filled in by double-clicking on each **Parameter**.

The screenshot shows the 'Set New CTI Connection Wizard Stage 8 of 16' window. The 'Active Recording' section is active, displaying 'Active Recording Interface Details'. A table lists parameters and their values, with the 'PrimaryAESMCCPort' row highlighted in blue. The 'Next' button is highlighted in red.

Parameter	Value
<b>PrimaryAESServerAddress</b>	
<b>PrimaryAESMCCPort</b>	4722
<b>PrimaryAESUserName</b>	
<b>PrimaryAESPassword</b>	
<b>PrimaryAESWarmStandby</b>	TRUE

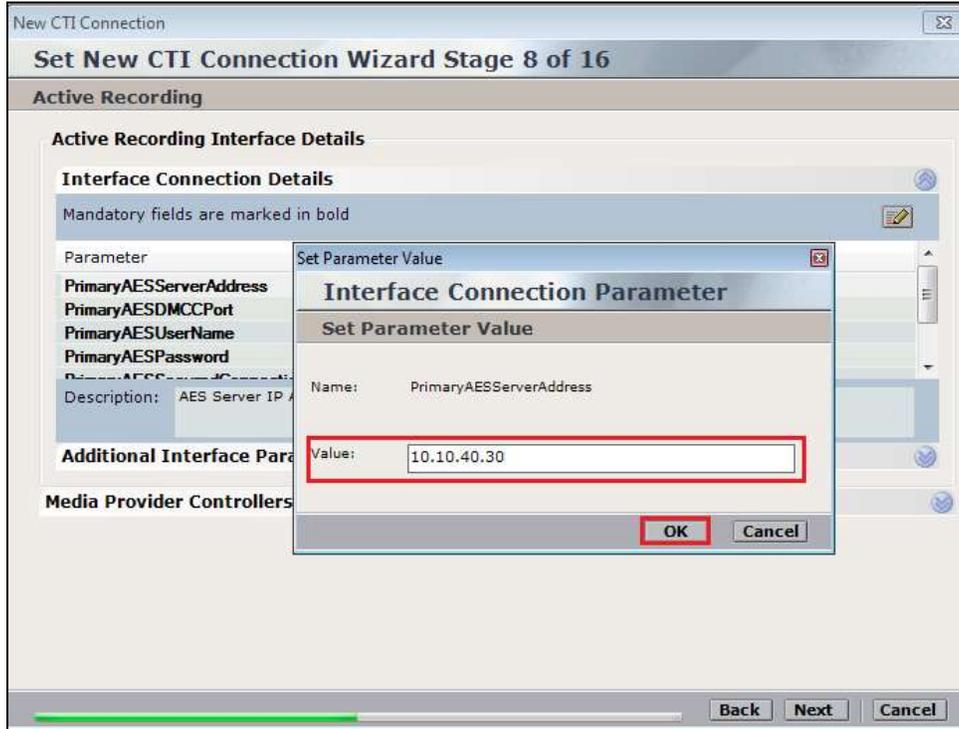
Description:

Additional Interface Parameters

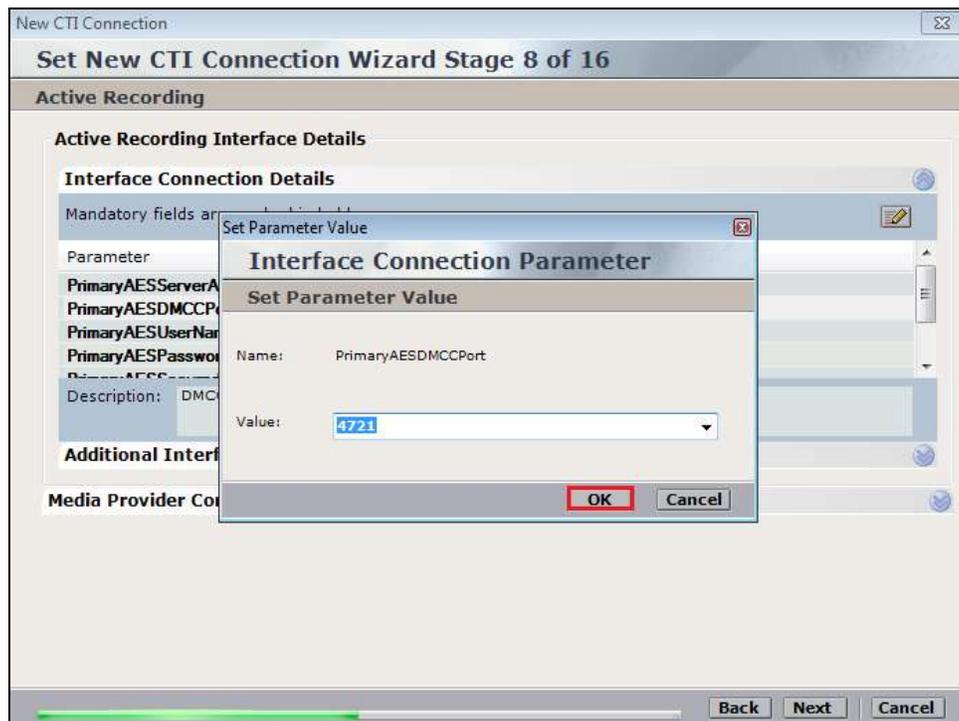
Media Provider Controllers - Location

Back Next Cancel

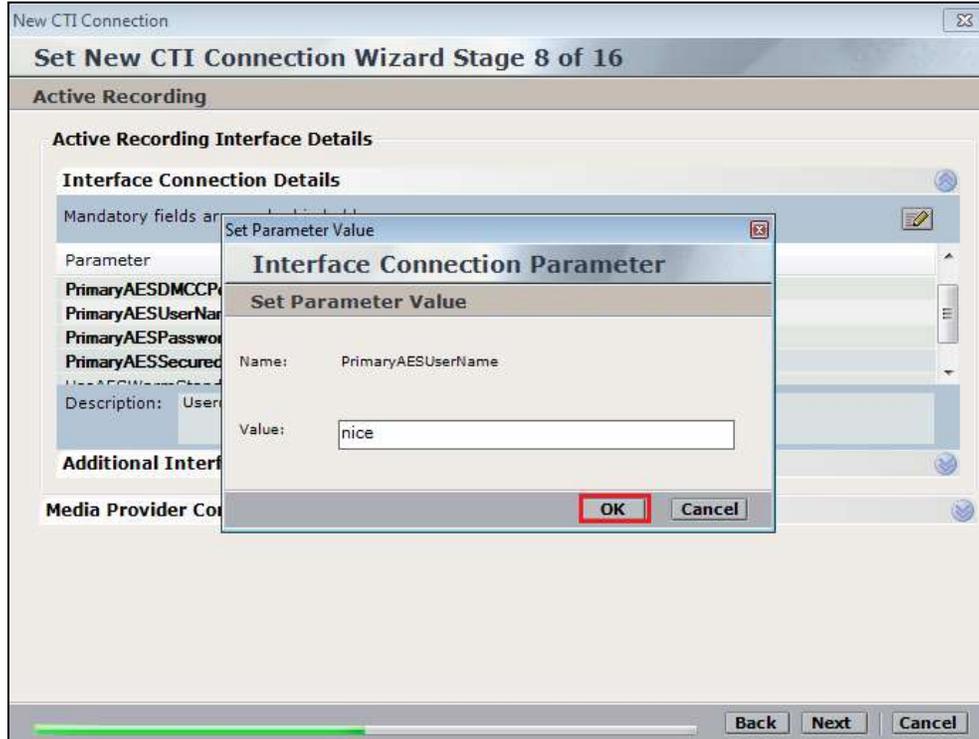
Enter the **Value** for the **AESServerAddress**. Click on **OK**.



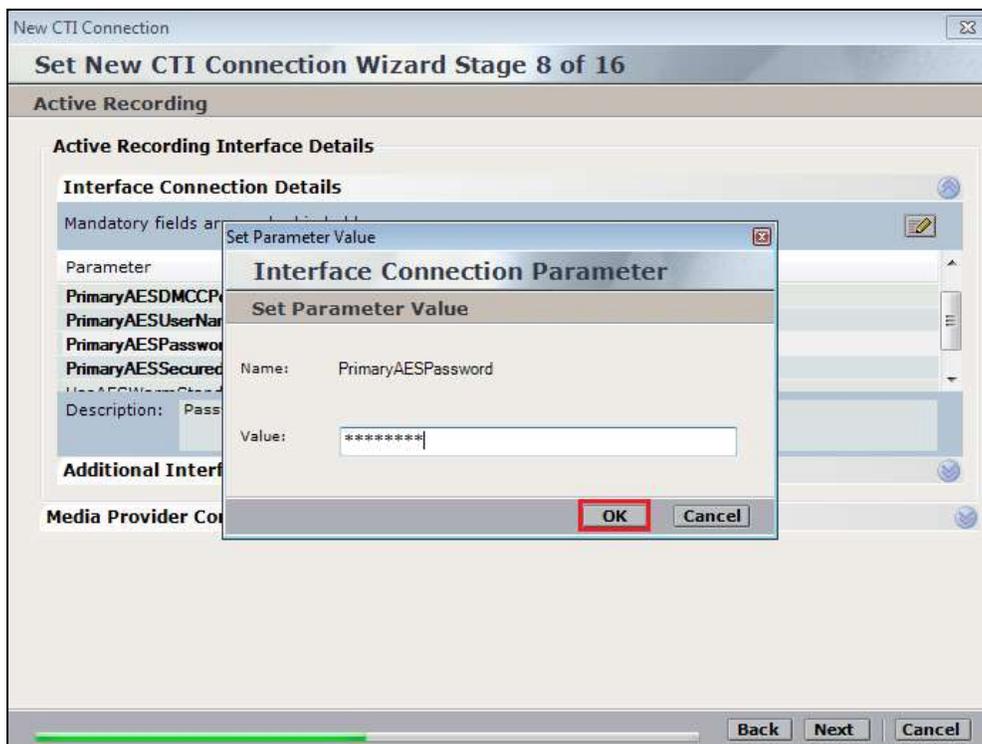
Enter the **Value** for the **AESDMCCPort**, note this will be the same port that was configured in **Section 6.5**. In this example the unencrypted port **4721** is entered.



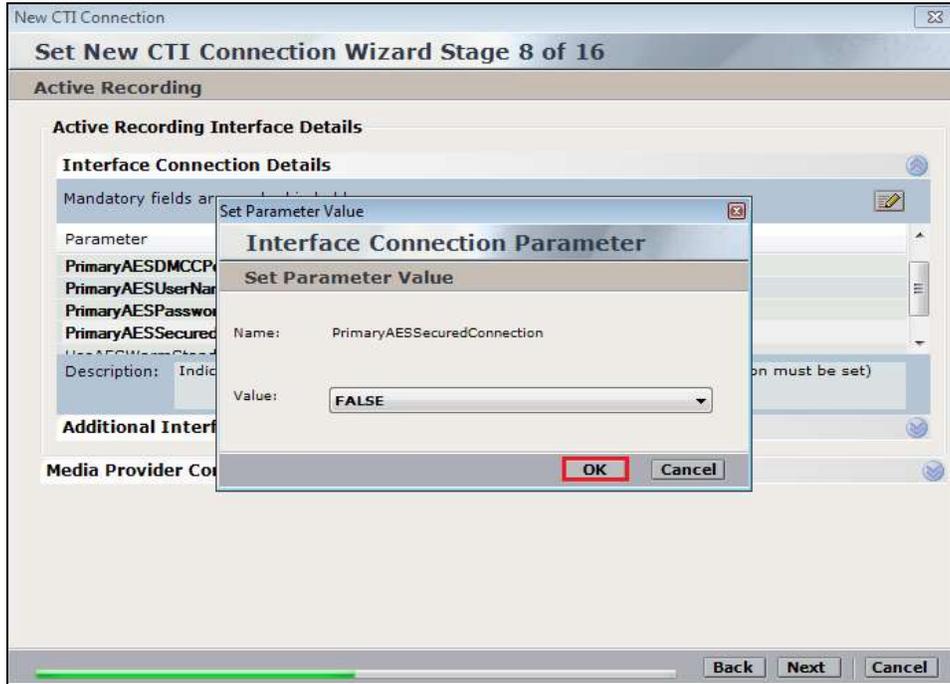
As before enter the username that was created in **Section 6.6** and click on **OK**.



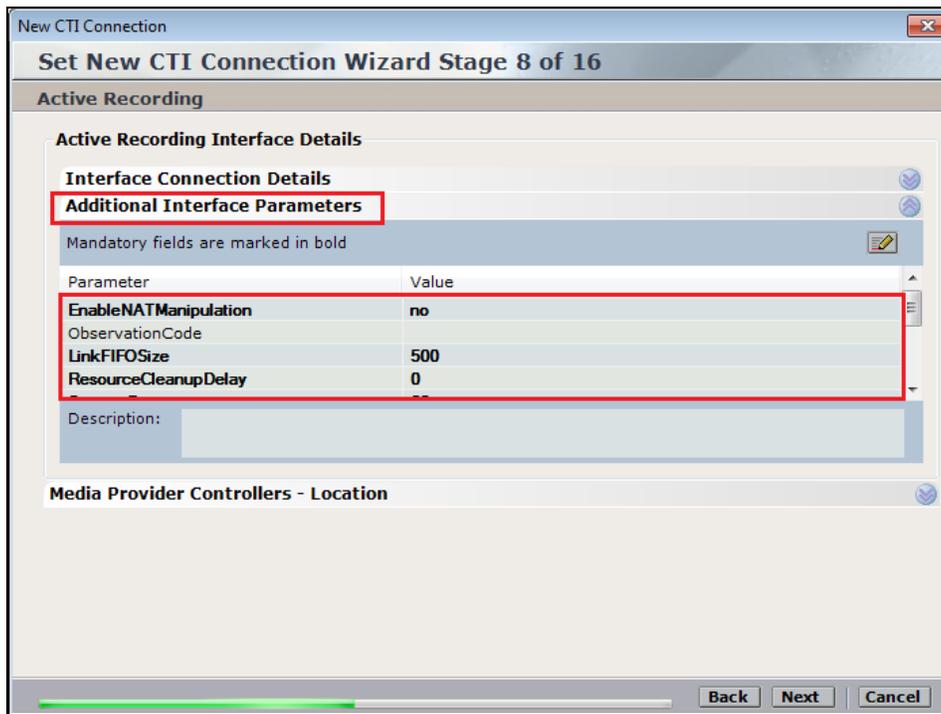
Enter the password that was created in **Section 6.6** and click on **OK**.



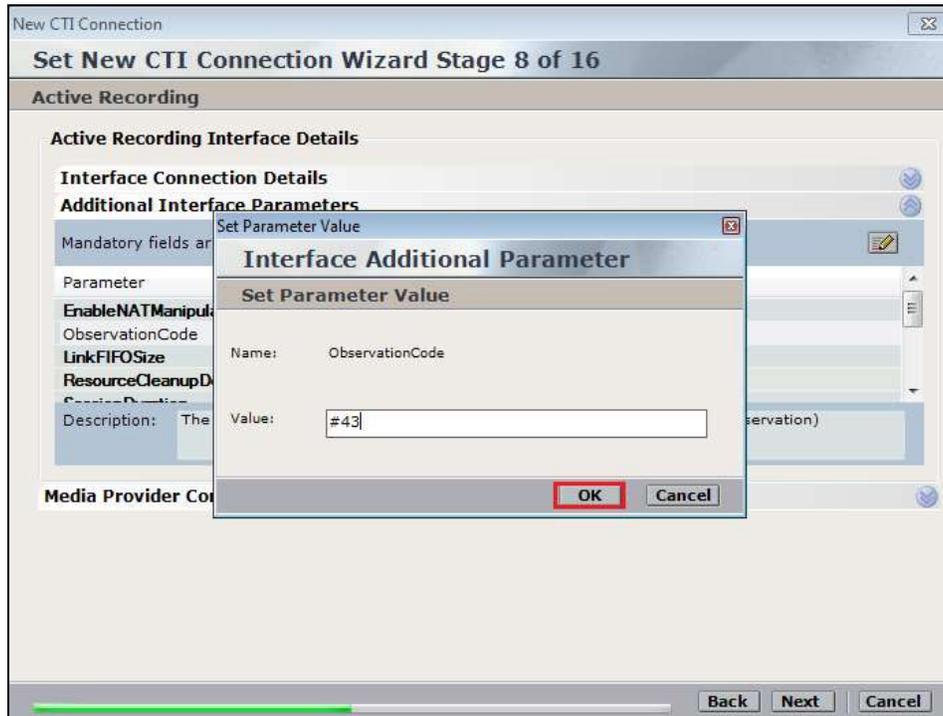
Because the unencrypted port was chosen select **False** for the **AESSecuredConnection**. Click on **OK** and then **Next** (not shown) to continue.



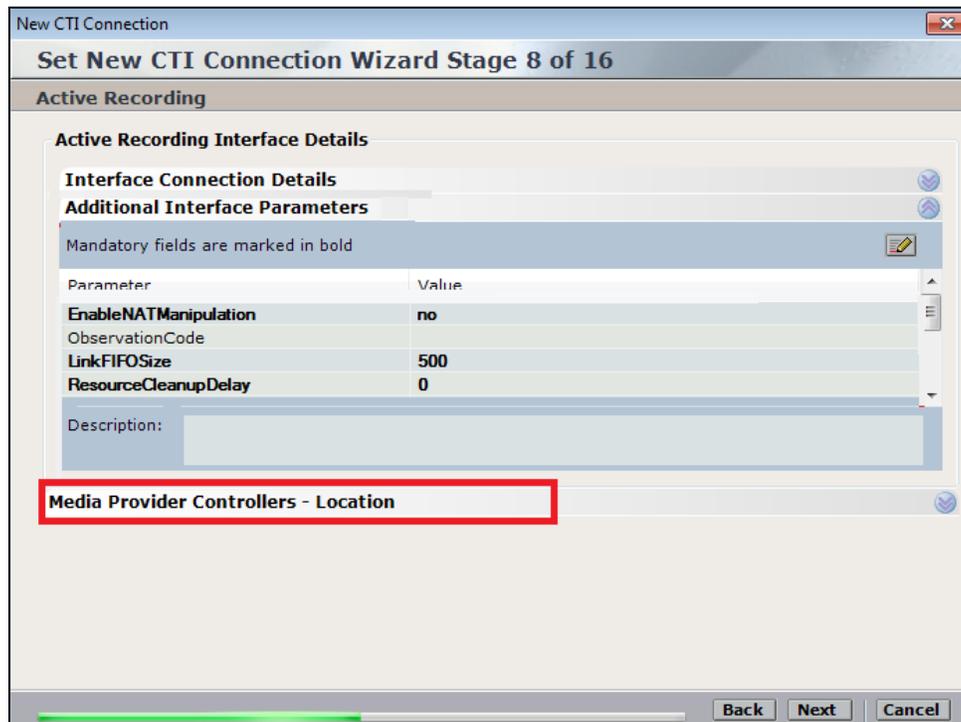
Click on **Additional Interface Parameters**, then to change the Service Observation Code double-click on **ObservationCode**.



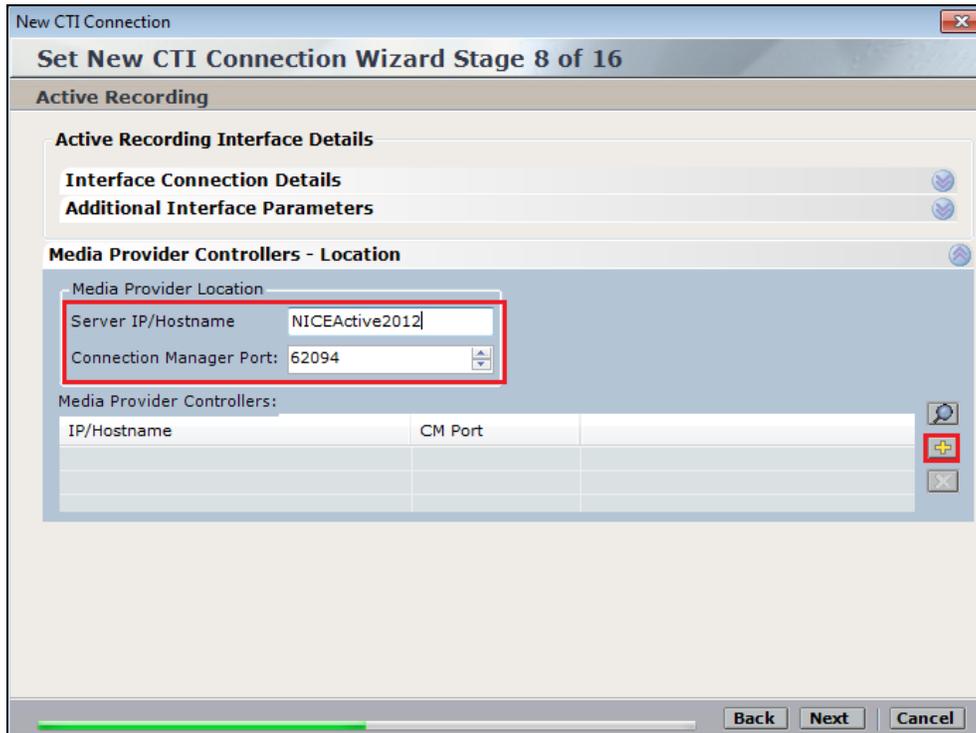
Enter the **Value** that was created in **Section 5.5**. This was the Service Observing Listen Only Access Code **#43**. Click on **OK** to continue.



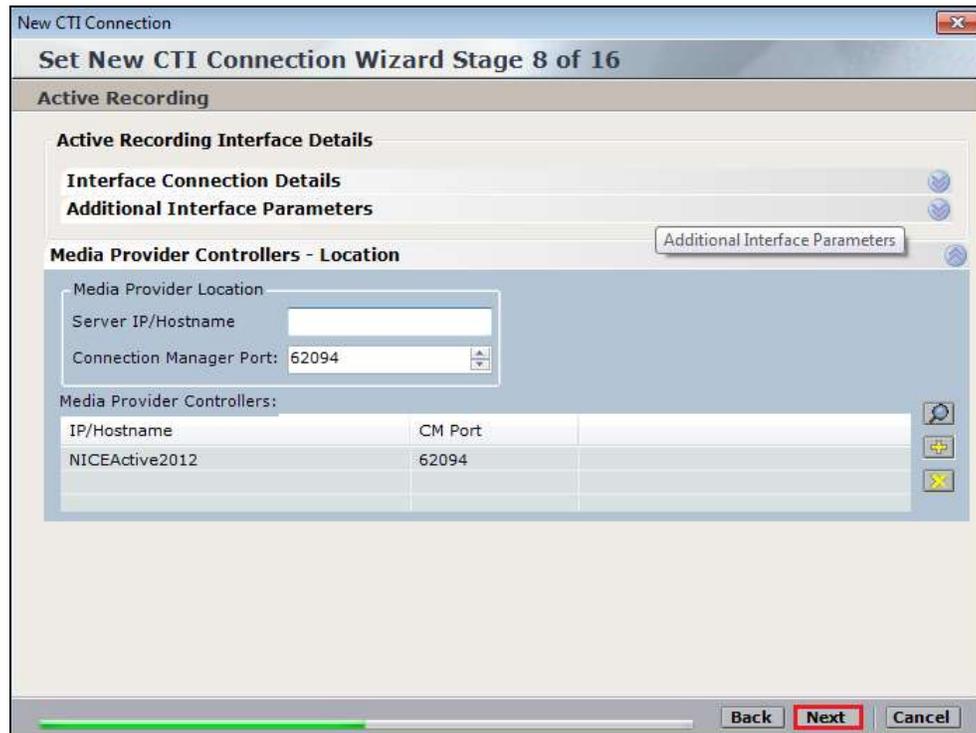
Click on **Media Provider Controllers – Location** to expand this.



Enter the **IP/Hostname** of the Nice Advanced Interactions Server, then click on the + icon to add this.



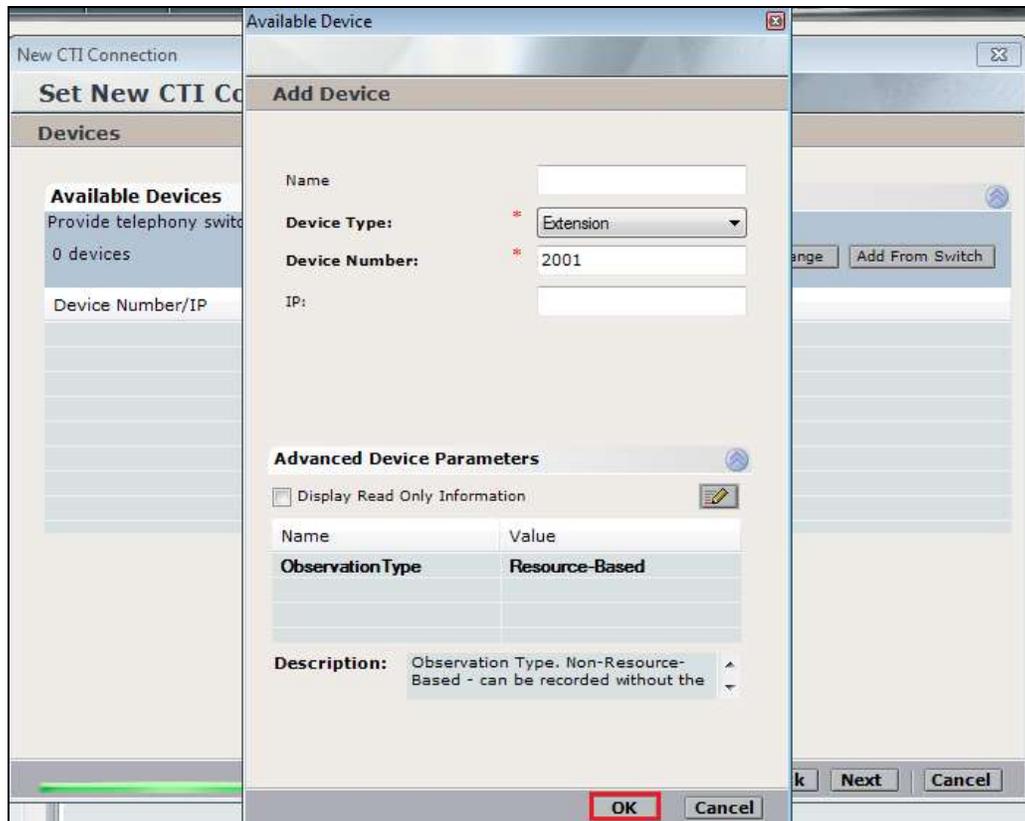
Click on **Next** to continue.



On the following screen, click on **Add**, to add the Communication Manager devices.



The **Device Type** should be **Extension** and insert the extension number of a phoneset that is to be recorded the example below showing extension **2001**. Expand **Advanced Device Parameters** and ensure that the **Value** for **Observation Type** is set to **Resourced-Based**. Click on **OK** to continue.



For Service Observe and Single Step Conference virtual extensions need to be added. These are the virtual extensions that were created in **Section 5.7**. Ensure that **Device Type** is set to **Virtual Extension** and add the correct extension for **Device Number**. Each of the **Parameters** highlighted at the bottom of the screen need to be configured and these are done by double-clicking on each parameter.

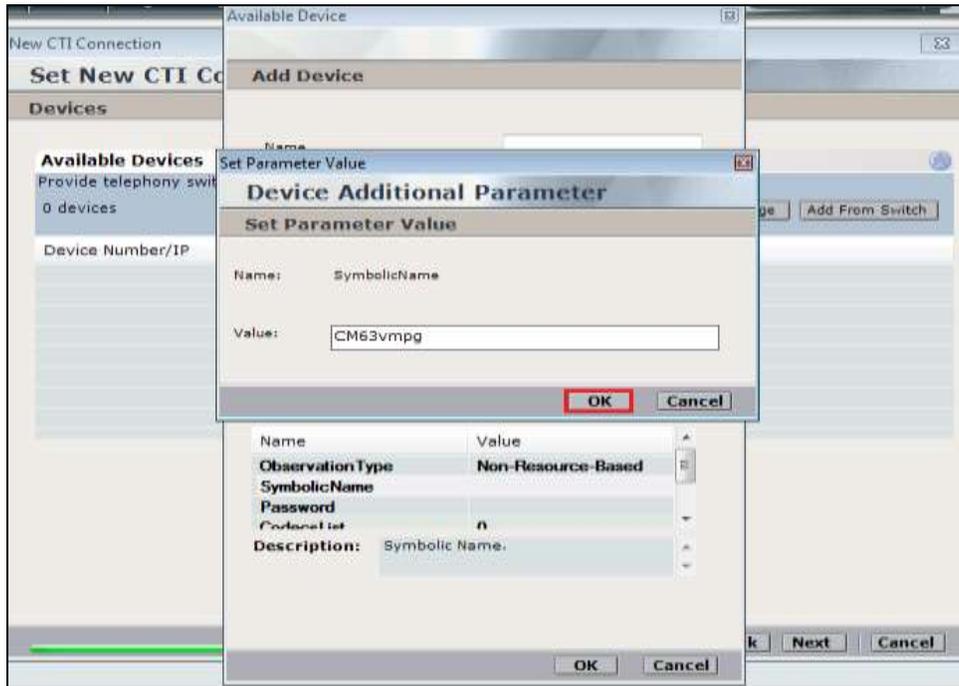
The screenshot shows a dialog box titled "Available Device" with a sub-header "Add Device". It contains several input fields and a table of parameters. A red box highlights the "Device Type" dropdown menu (set to "Virtual Extension") and the "Device Number" text box (containing "28902"). Another red box highlights a table of "Advanced Device Parameters" with the "ObservationType" row selected. The "Description" field is also visible below the table.

**Advanced Device Parameters**

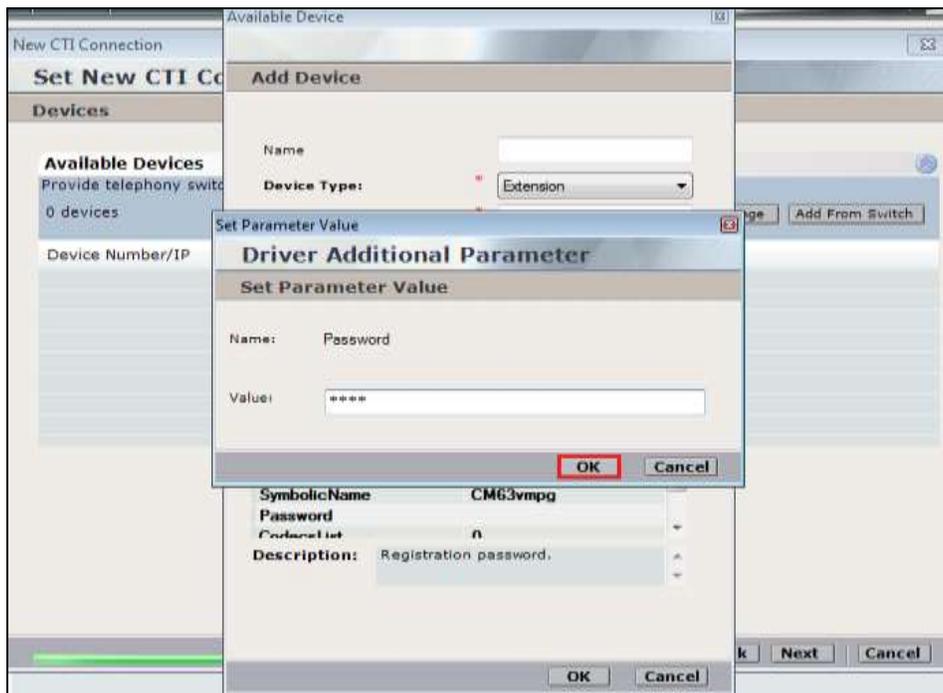
Name	Value
ObservationType	None
SymbolicName	
Password	
ControlList	n

**Description:** Observation Type. Non-Resource-Based - can be recorded without the

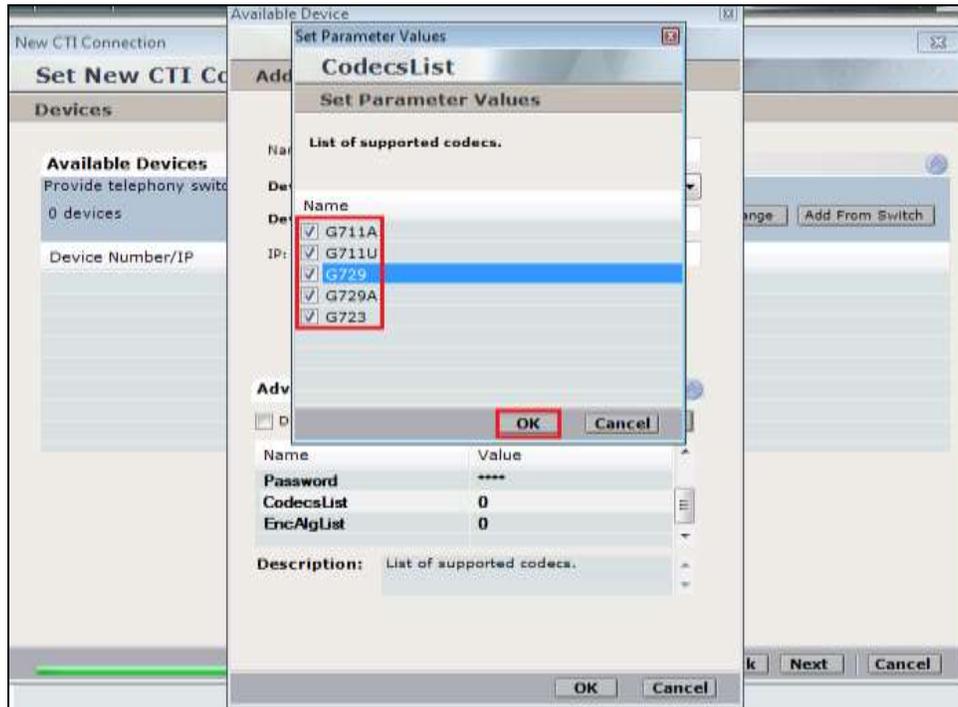
Enter the correct **Value** for **SymbolicName**. Double-click on **SymbolicName** to set the value. This should be the same as the switch name entered in **Section 6.2**.



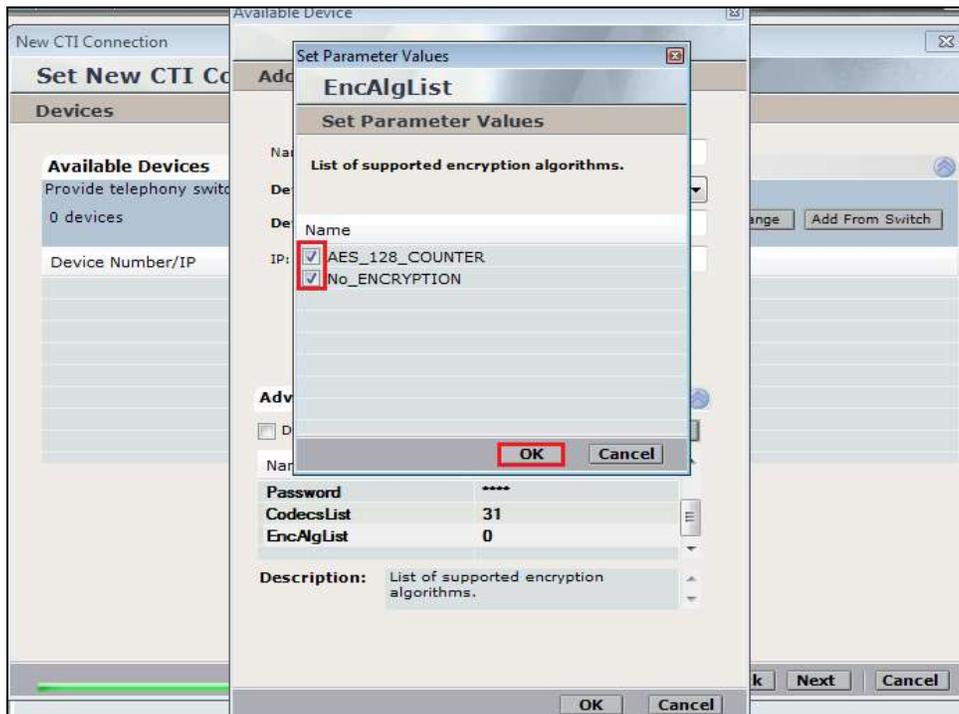
Enter the correct **Password** and note this is the password for the extension that is being added here. This is the station password which was entered during the creation of the station. A printout of an extension can be found in **Section 5.5** of these Application Notes.



Double-click on **CodecsList** and ensure that all the values are ticked as shown below. Click on **OK** to continue.



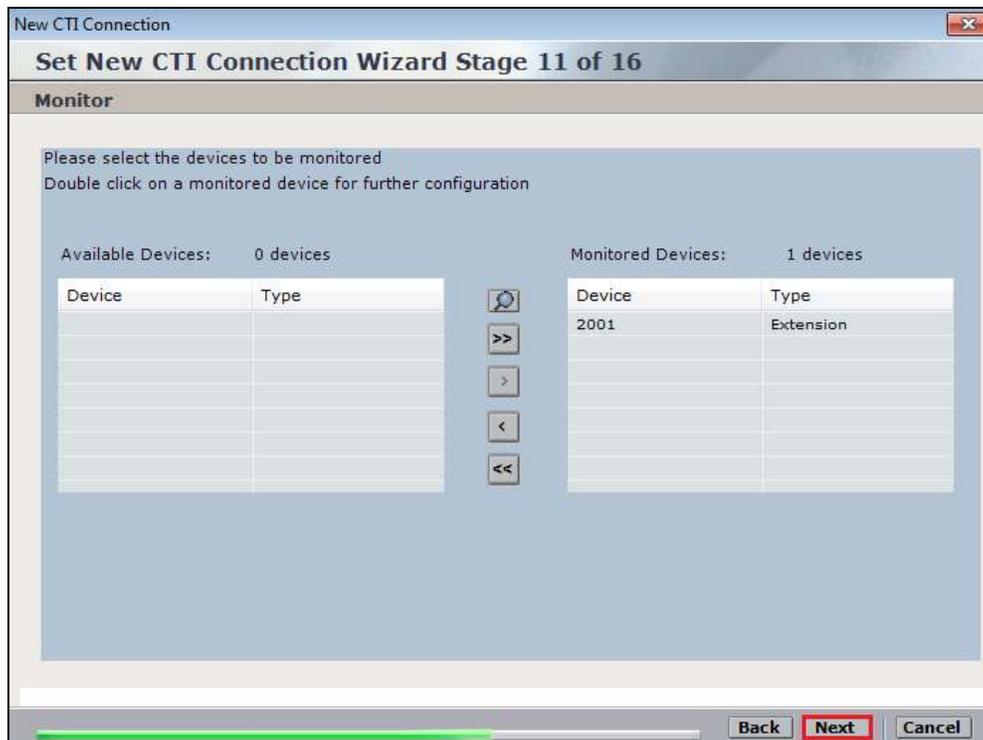
Double-click on **EncAlgList** and ensure both options are ticked as shown below. Click on **OK** to continue.



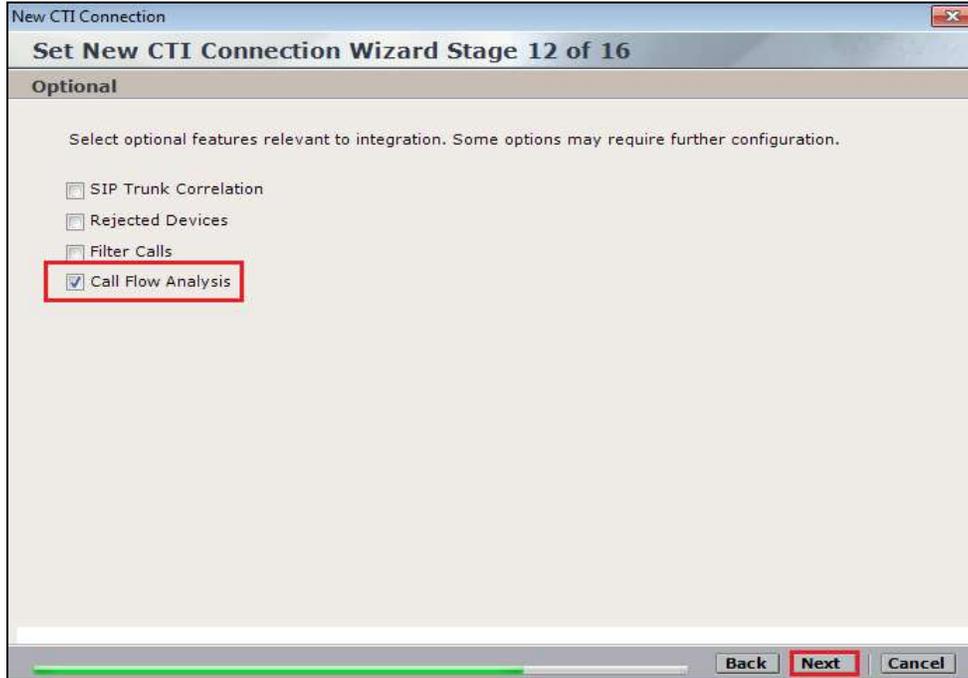
Click on **Next** to continue.



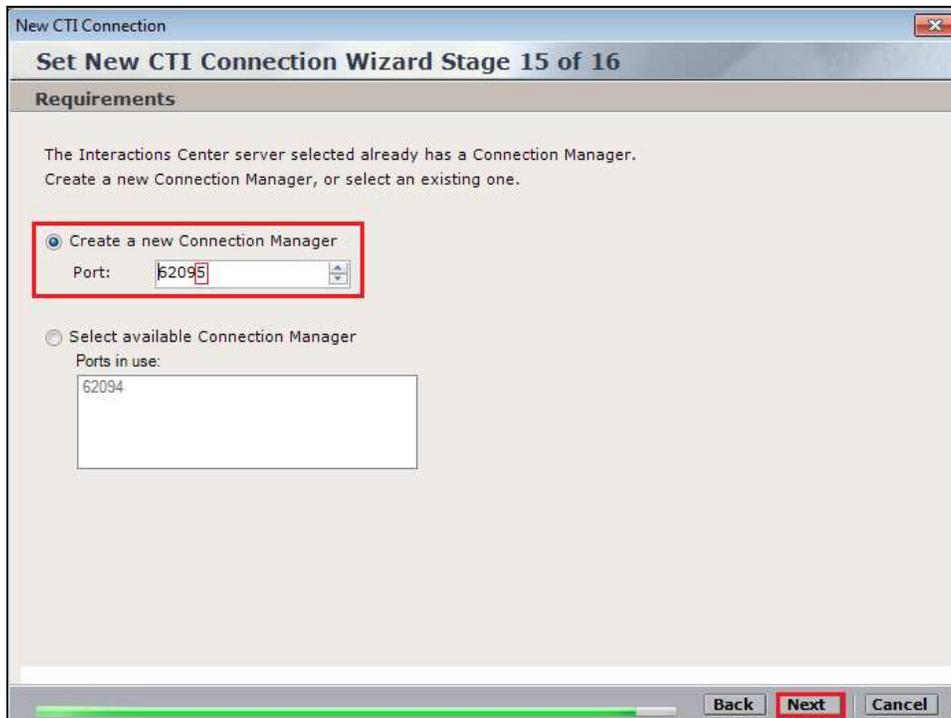
Select the new extension and click on the >> icon as shown. Click on **Next** to continue.



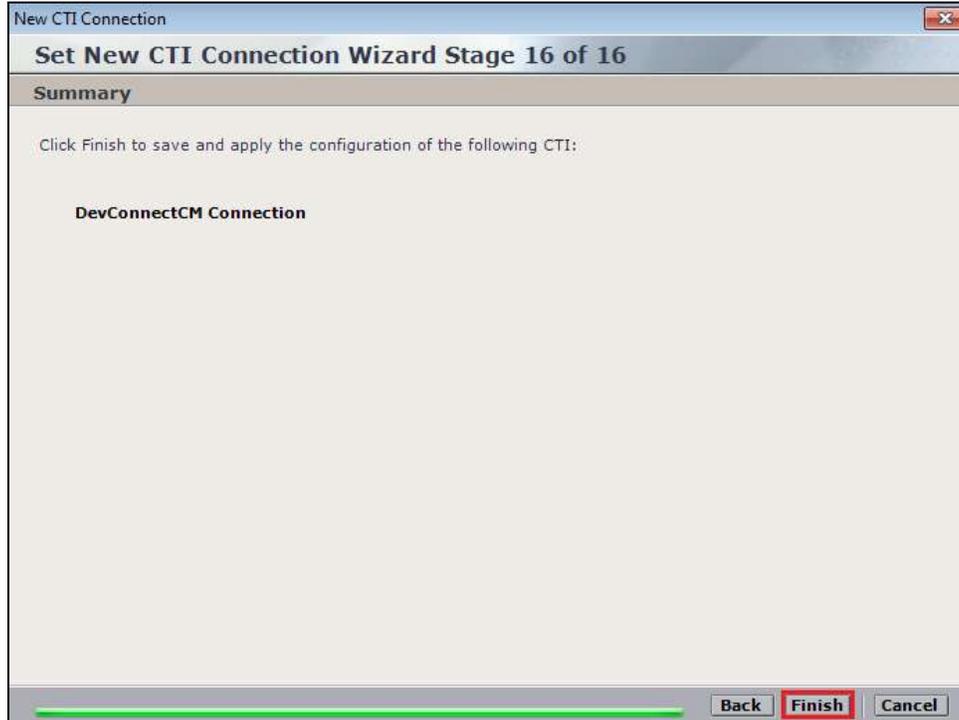
It is optional, but for better analysis tick on **Call Flow Analysis** and click on **Next** to continue.



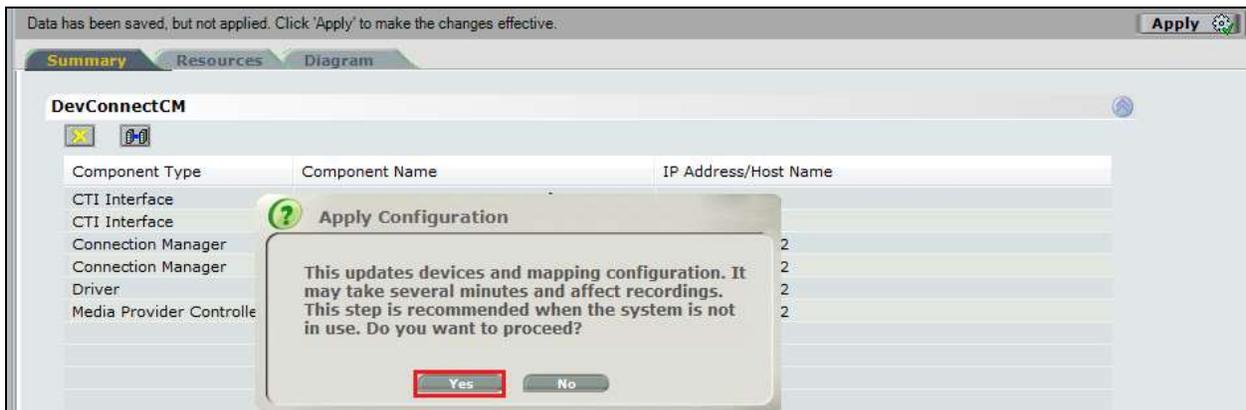
Select a different **Port** number as shown below **62095** is chosen simply because **62094** was already in use.



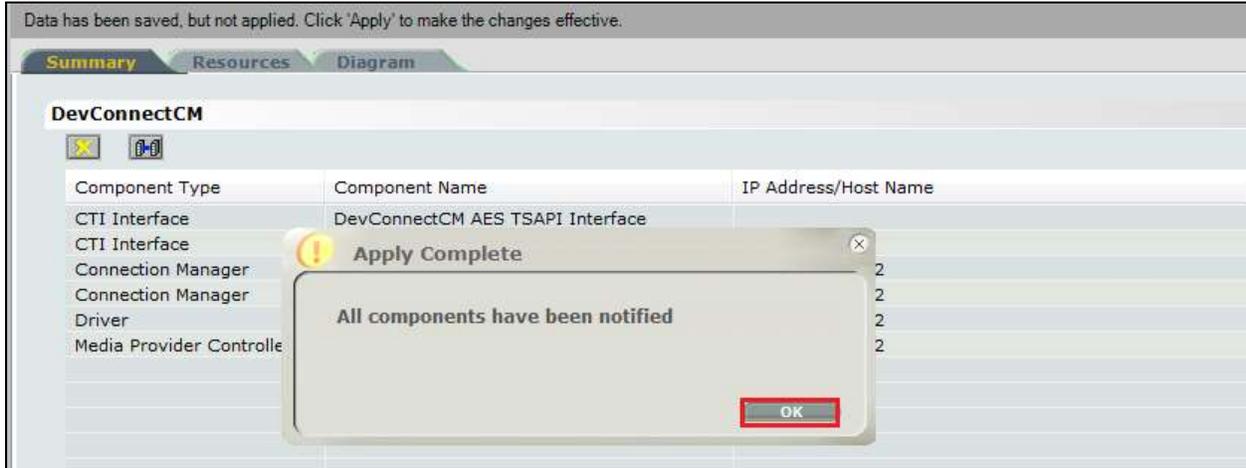
Click on **Finish** to complete the **New CTI Wizard**.



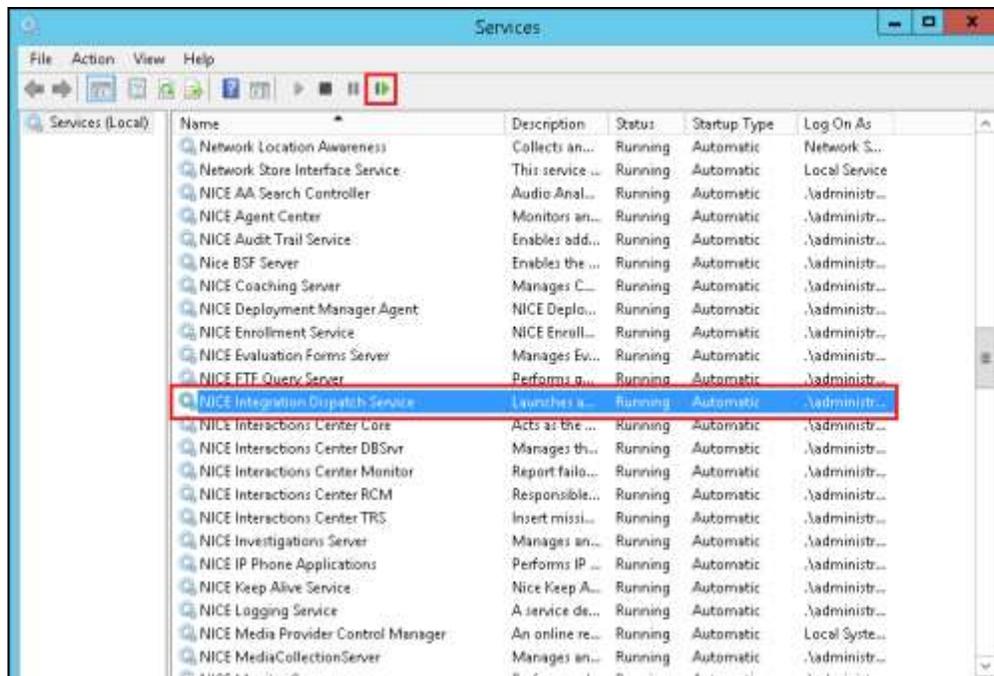
Click on **Apply** at the top right of the screen to save the new connection and click on **Yes** to proceed



The following shows that the save was successful. Click on **OK** to continue.

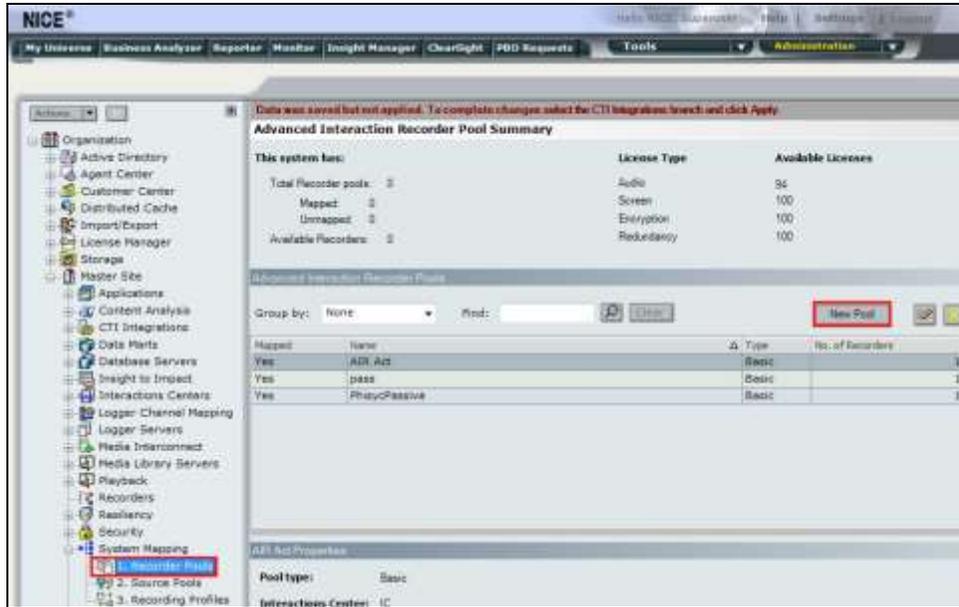


From the NICE Application Server, open **Services** and restart the **NICE Integration Dispatch Service**.

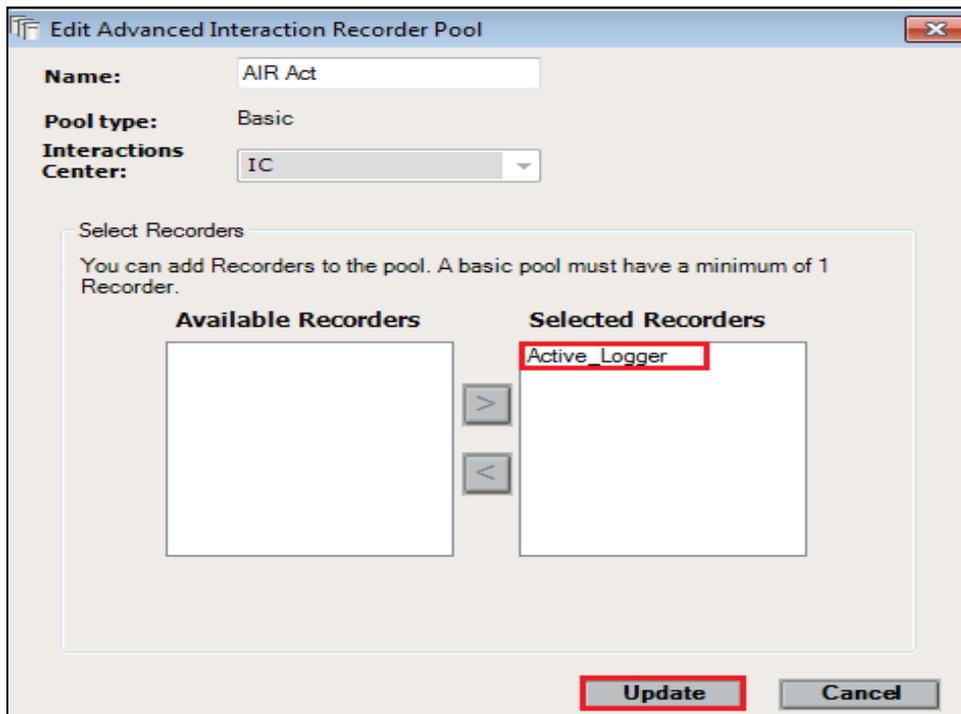


## 7.2. System Mapping

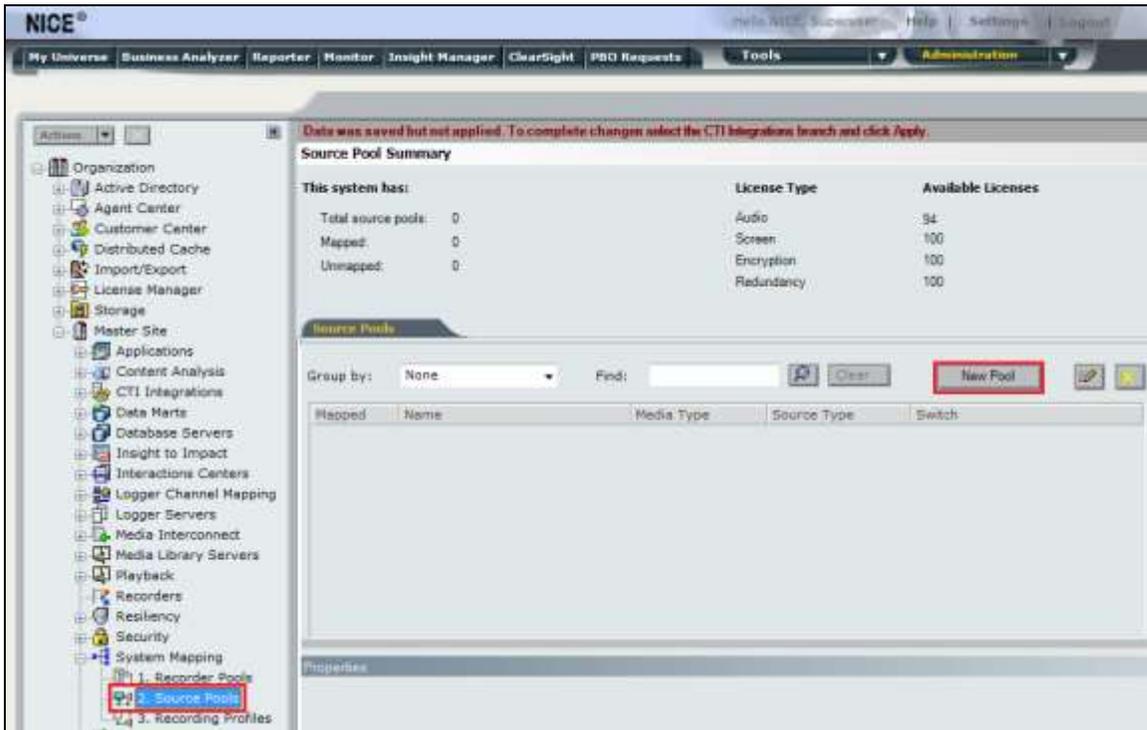
From the web browser navigate to **Master Site** → **System Mapping** → **Recorder Pools**. In the main window click on **New Pool**.



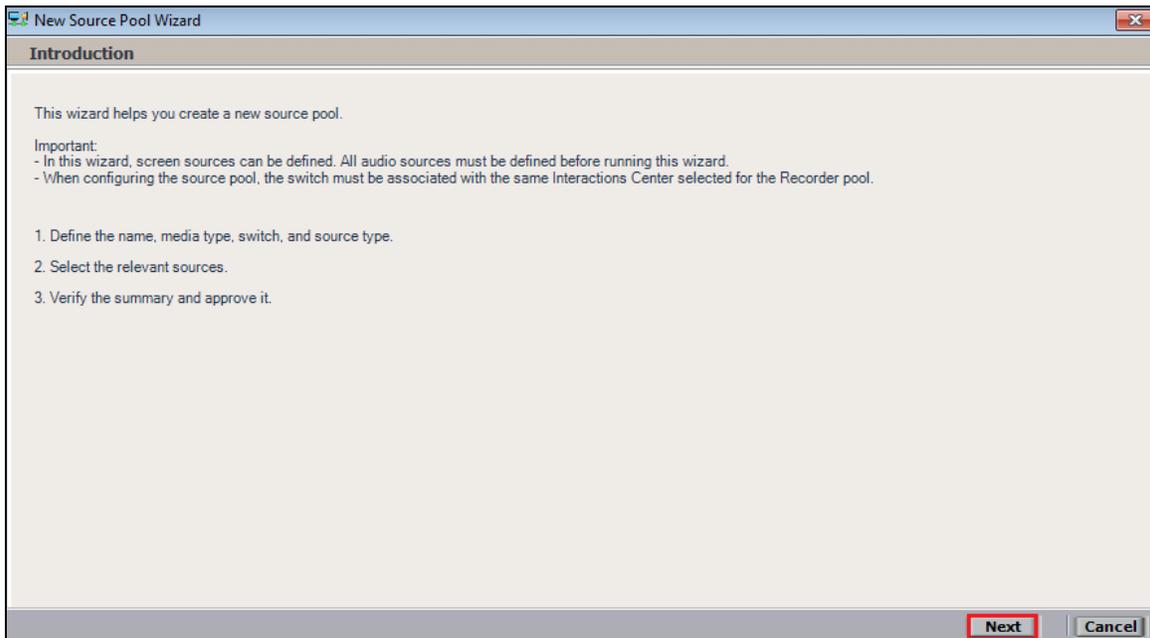
Enter a suitable **Name** for the **Recorder Pool** and select the **Active\_Logger** from the list of **Available Recorders** and click on **Update** to continue.



From the left navigation window select **Source Pools** and from the main window click on **New Pool**.



Click on **Next** to continue to add a new **Source Pool**.



Enter a suitable **Name** and the other values were left as default. Click on **Next** to continue.

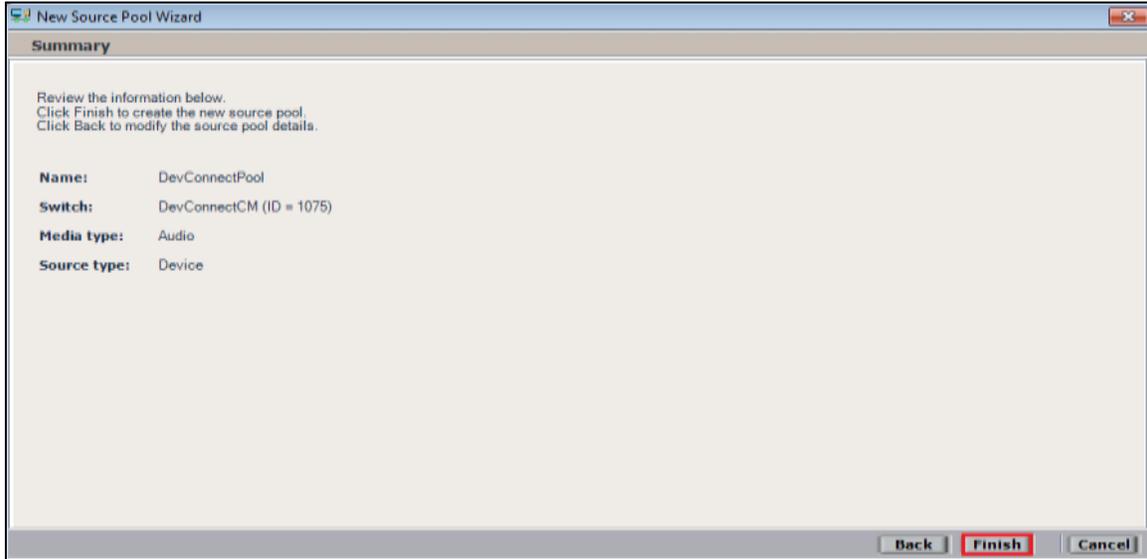
The screenshot shows the 'Define Source Pool' step of the 'New Source Pool Wizard'. The window title is 'New Source Pool Wizard' and the subtitle is 'Define Source Pool'. Below the subtitle is the instruction: 'Define the source pool details. After completing this wizard, the media type, switch, and source type cannot be changed.' There are four input fields: 'Name' with the value 'DevConnectPool', 'Media type' with the value 'Audio', 'Switch' with the value 'DevConnectCM (ID = 1075)', and 'Source type' with the value 'Device'. At the bottom right, there are three buttons: 'Back', 'Next', and 'Cancel'. The 'Next' button is highlighted with a red box.

Select the extensions that were created in **Section 7.1**, note only one extension number is shown in the example below but this is not typical. Click on **Next** to continue.

The screenshot shows the 'Select Sources' step of the 'New Source Pool Wizard'. The window title is 'New Source Pool Wizard' and the subtitle is 'Select Sources'. At the top left, there is a 'Find:' search box with a magnifying glass icon and a 'Clear' button. At the top right, it says 'Selected: 1/1' with 'Select All' and 'Clear Selection' buttons. Below this is a table with the following columns: 'Name', 'Device Number', 'Unique Device ID', and 'IP Address'. The first row has a checked checkbox in the 'Name' column, and the 'Device Number' is '2001'. At the bottom right, there are three buttons: 'Back', 'Next', and 'Cancel'. The 'Next' button is highlighted with a red box.

Name	Device Number	Unique Device ID	IP Address
<input checked="" type="checkbox"/>	2001		

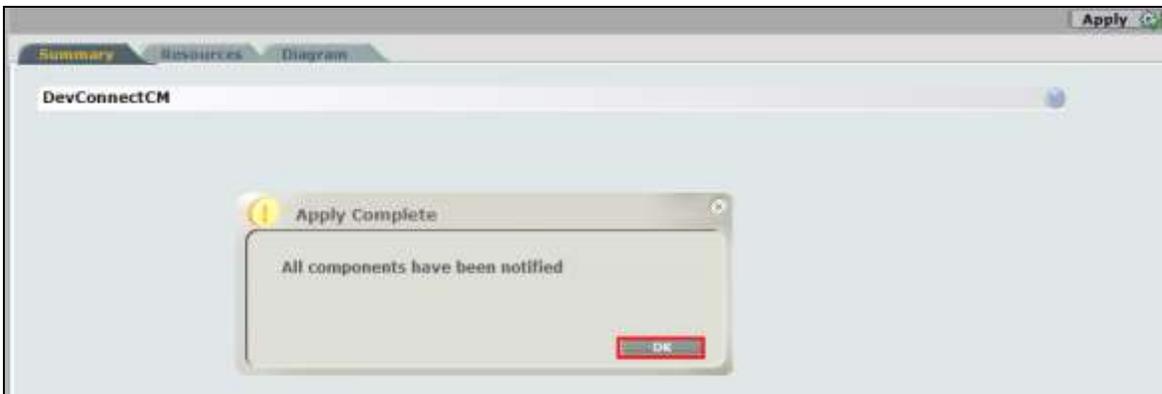
Click on **Finish** to complete the **New Source Pool Wizard**.



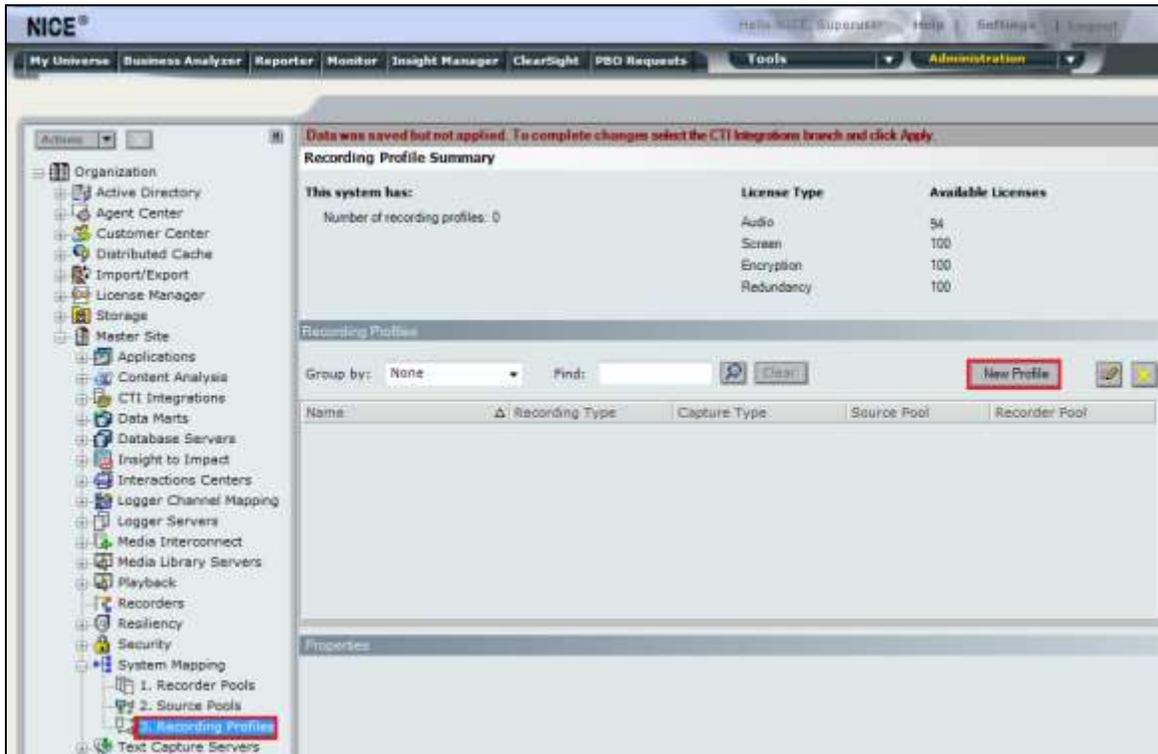
To implement these new changes, navigate to **Master Site** → **CTI Integrations** in the left window and in the main window click on **Apply** at the top right of the window.



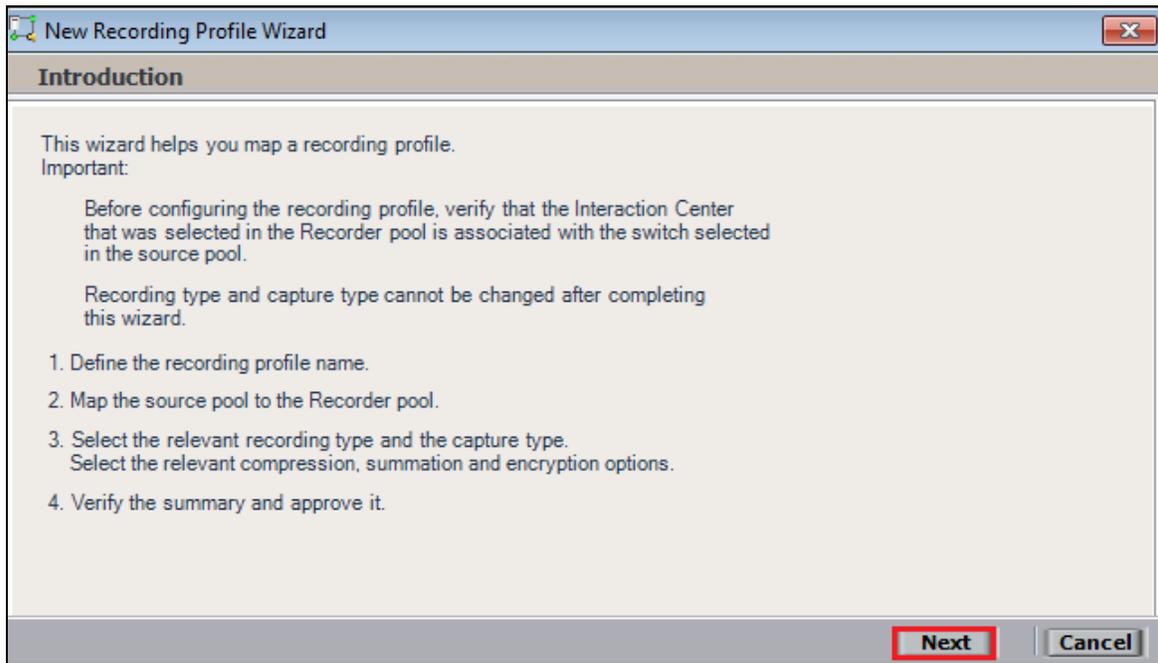
The following screen shows the changes were saved correctly. Click on **OK** to continue.



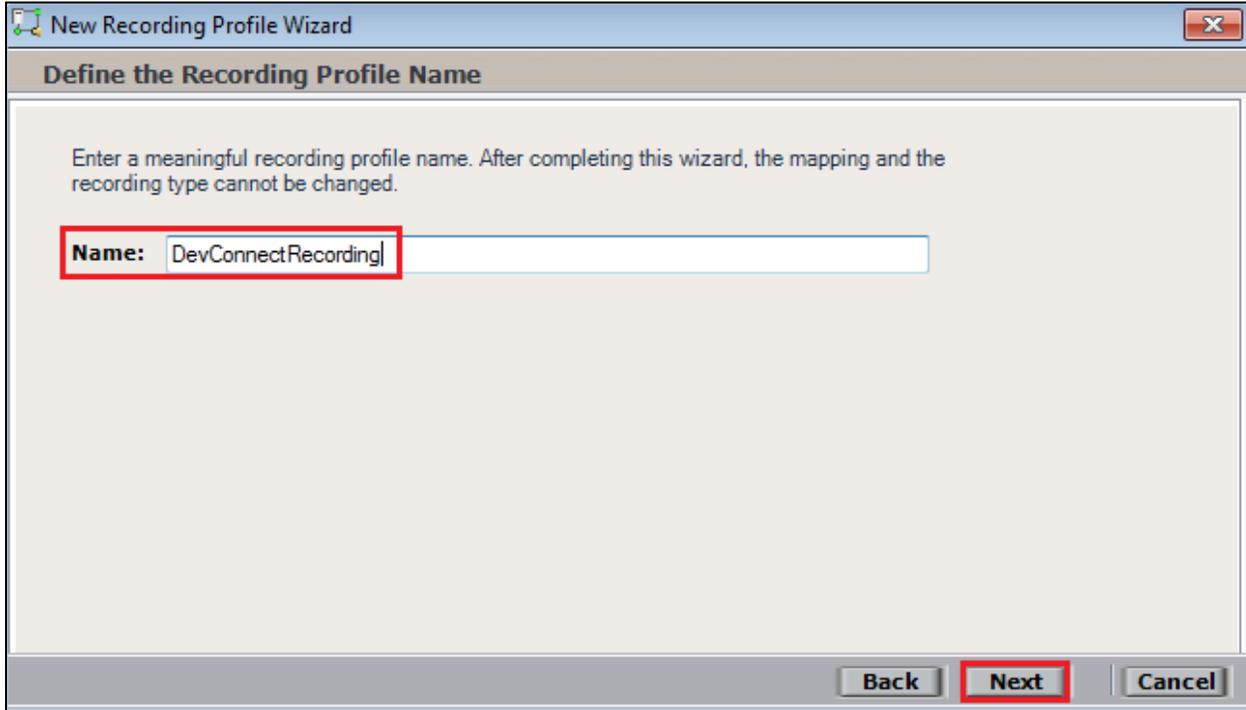
From the left window navigate to **Master Site** → **System Mapping** → **Recording Profiles** and in the main window click on **New Profile**.



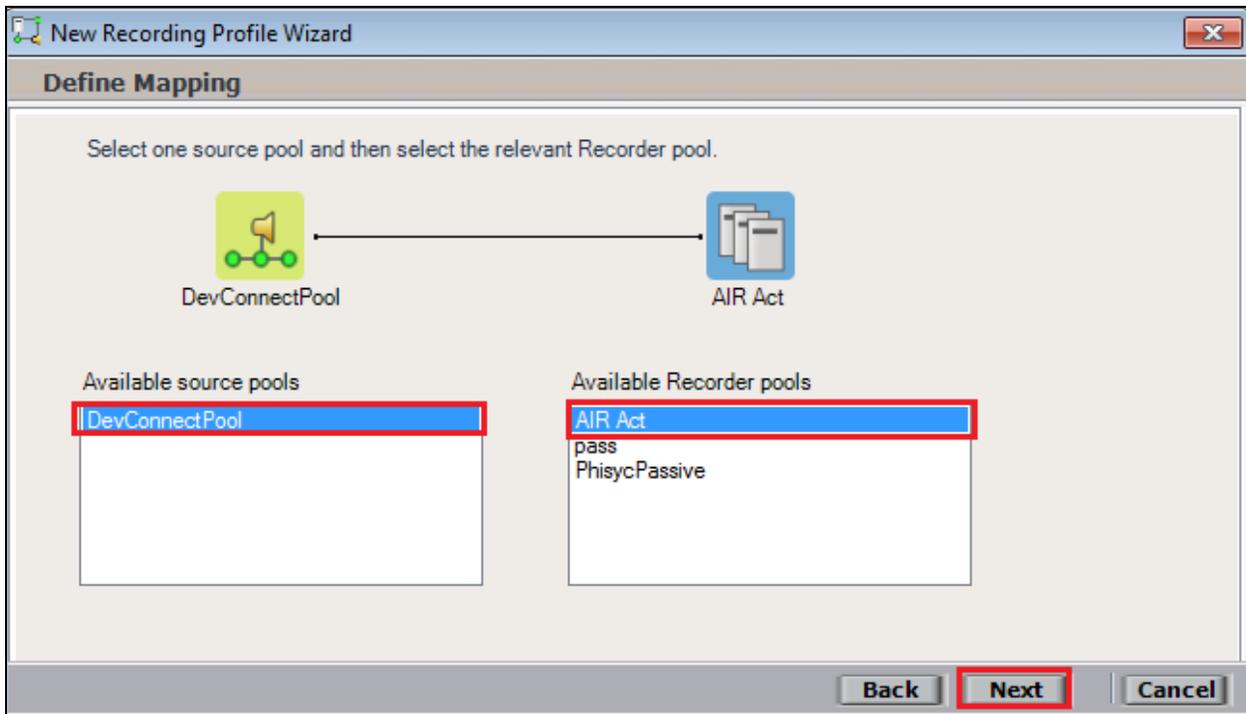
Click on **Next** to continue with the **New Recording Profile Wizard**.



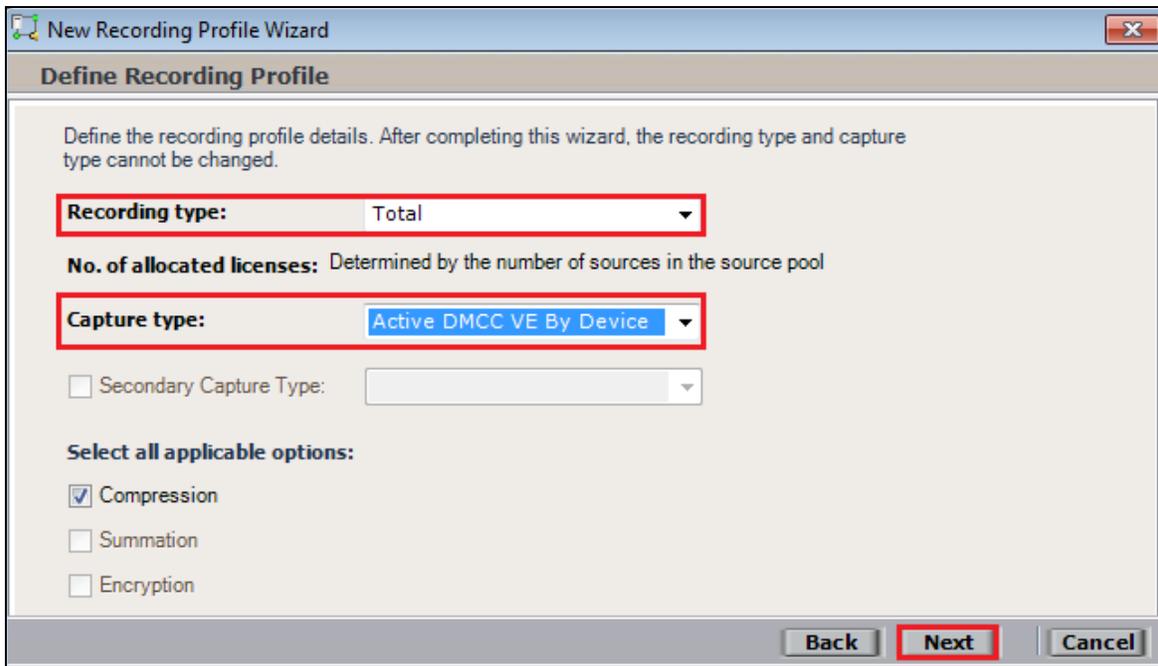
Enter a suitable **Name** for the Recording profile.



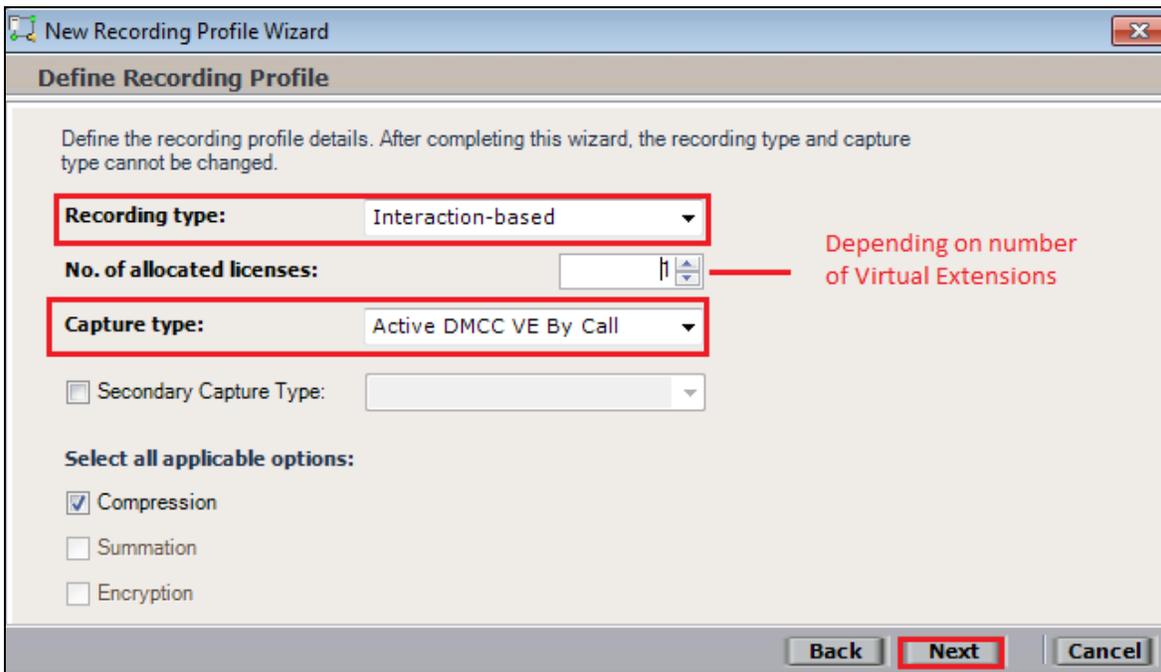
Select the correct **source pool** and **Recorder pool**, and then click **Next** to continue.



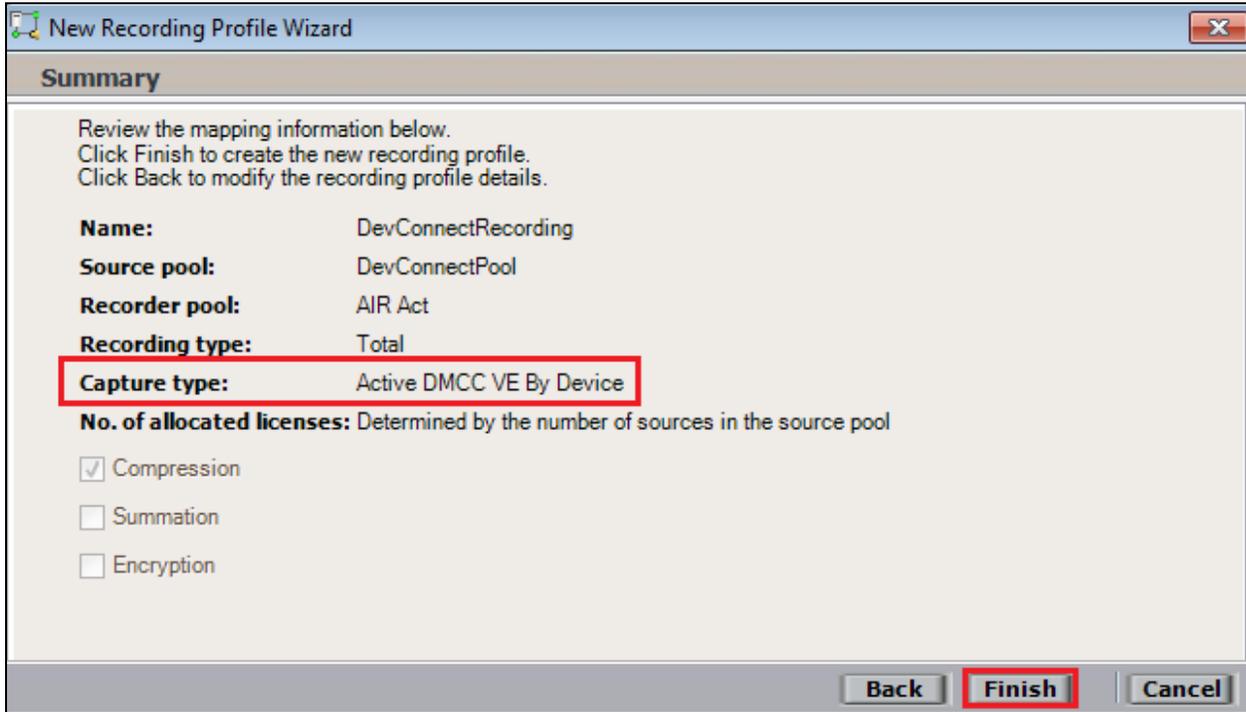
For total recording i.e., the recording of all calls, select **Total** as the **Recording type**. For **Capture type** ensure that **Active DMCC MR** is selected from the drop-down box. Compression is selected as default and can be left like this. Click on **Next** to continue.



**Note:** The only difference in the setup for Single Step Conference is with both the choice of **Recording type** which is set to **Interaction-based** and **Capture type** which will be **Active DMCC VE By Call** as shown below.



Click on **Finish** to complete the **New Recording Profile Wizard**. The screen below shows that for Service Observe.



Navigate to **Master Site** → **CTI Integrations** and from the main window click on **Apply**. Then click on **Yes** to proceed.



This concludes the setup of the NICE Application Server for DMCC Service Observe and Single Step Conference recording.

## 8. Verification Steps

This section provides the steps that can be taken to verify correct configuration of the NICE Engage Platform and Avaya Aura® Application Enablement Services.

### 8.1. Verify Avaya Aura® Communication Manager CTI Service State

Before the connection between the NICE Engage Platform and the AES is check the connection between Communication Manager and AES can be check to ensure it is functioning correctly. Check the AESVCS link status by using the command **status aesvcs cti-link**. Verify the **Service State** of the CTI link is **established**.

```
status aesvcs cti-link
```

AE SERVICES CTI LINK STATUS						
CTI Link	Version	Mnt Busy	AE Services Server	Service State	Msgs Sent	Msgs Rcvd
1	4	no	aes63vmpg	<b>established</b>	18	18

### 8.2. Verify TSAPI Link

On the AES Management Console verify the status of the TSAPI link by selecting **Status** → **Status and Control** → **TSAPI Service Summary** to display the **TSAPI Link Details** screen. Verify the status of the TSAPI link by checking that the **Status** is **Talking** and the **State** is **Online**.

AVAYA Application Enablement Services Management Console

Welcome! User: cm6t  
Last login: Thu Feb 20 11:01:32 2014 from 192.168.10.223  
Number of prior failed login attempts: 33  
hostname: AES63VMPG  
Server Offer Type: VIRTUAL\_APPLIANCE\_OVF\_VMWARE  
SW Version: 6.3.0.0.213-0  
Server Date and Time: Thu Feb 20 11:14:02 UTC 2014

Status | Status and Control | TSAPI Service Summary Home | Help | Logout

AE Services  
Communication Manager Interface  
Licensing  
Maintenance  
Networking  
Security  
\* Status  
Alarm Viewer  
Log Manager  
Logs  
\* Status and Control  
  • CxLAN Service Summary  
  • DLG Services Summary  
  • DMCC Service Summary  
  • Switch Conn Summary  
  • TSAPI Service Summary

TSAPI Link Details

Enable page refresh every 60 seconds

Link	Switch Name	Switch CTI Link ID	Status	Since	State	Switch Version	Associations	Hops to Switch	Hops from Switch	Hops Period
1	CH63vmpg	1	Talking	Tue Feb 18 11:21:49 2014	Online	16	5	15	15	30

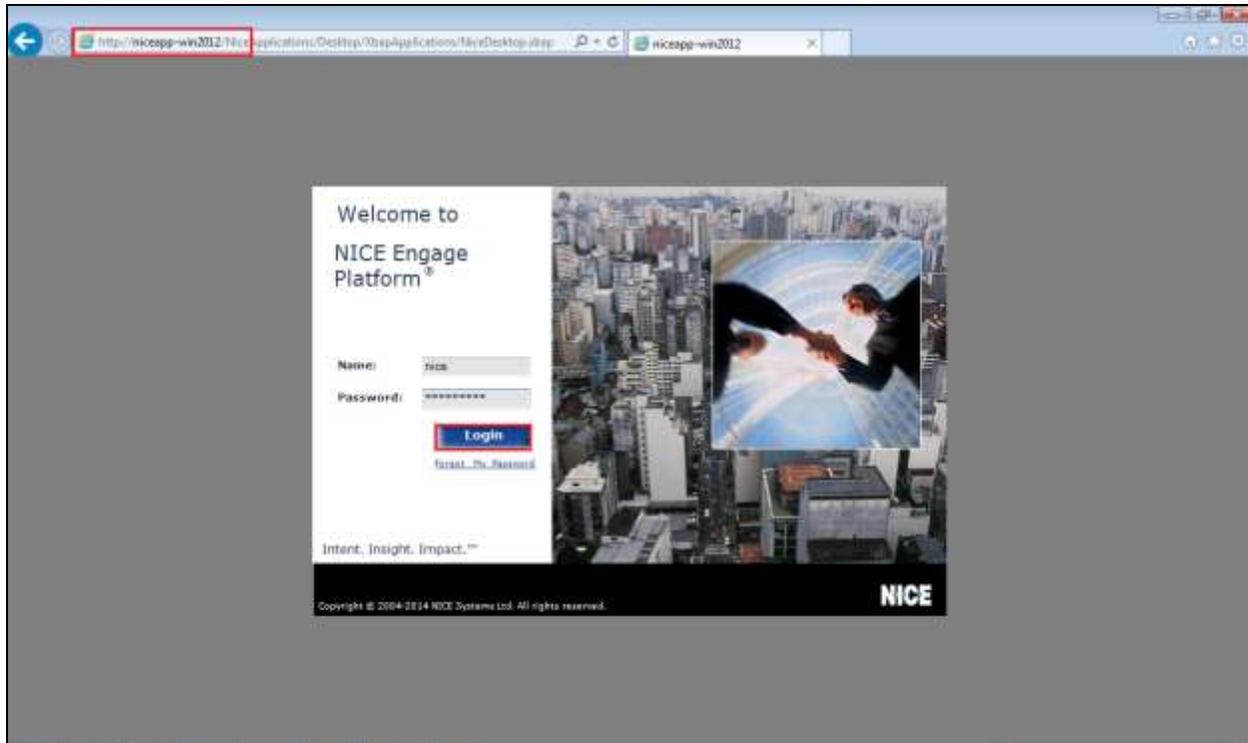
Online Offline

For service-wide information, choose one of the following:

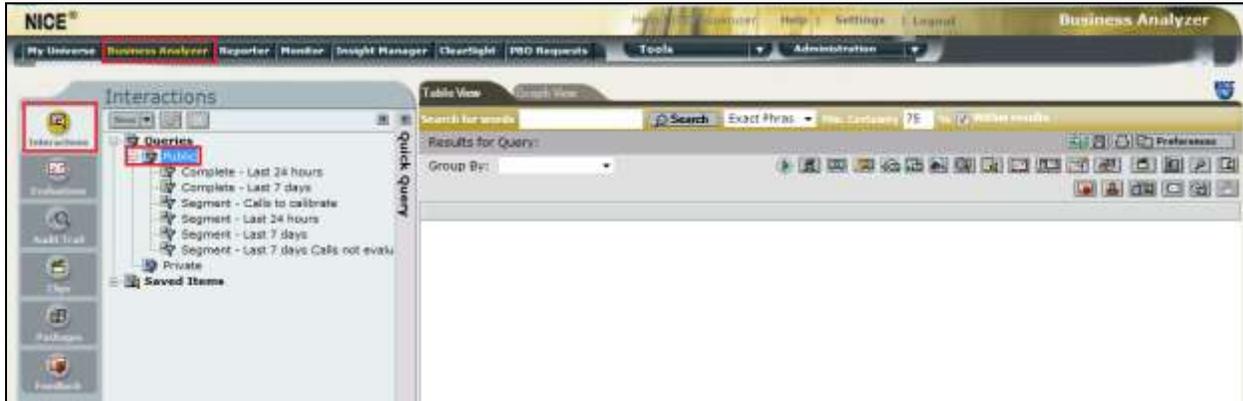
### 8.3. Verify calls are being recorded

From any of the monitored Avaya endpoints make a series of inbound and outbound calls. Once these calls are completed they should be available for playback through a web browser to the NICE Application Server.

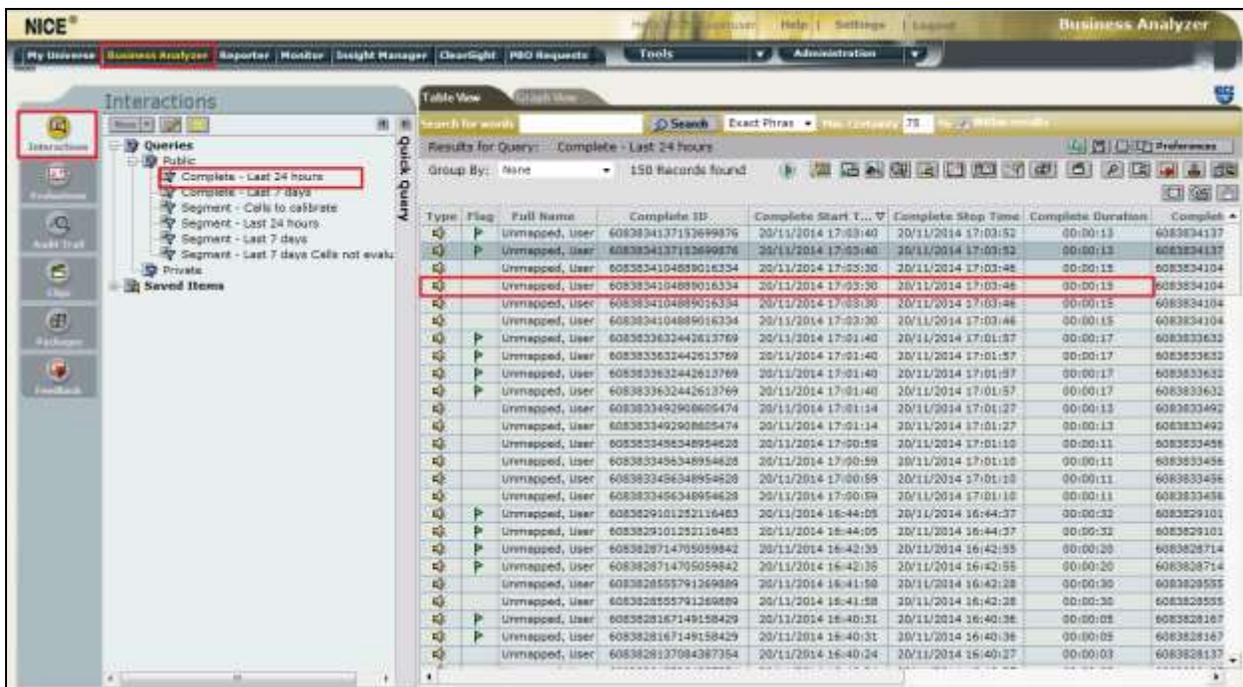
Open a browser session to the NICE Application Server as is shown below. Enter the proper credentials and click on **Login**.



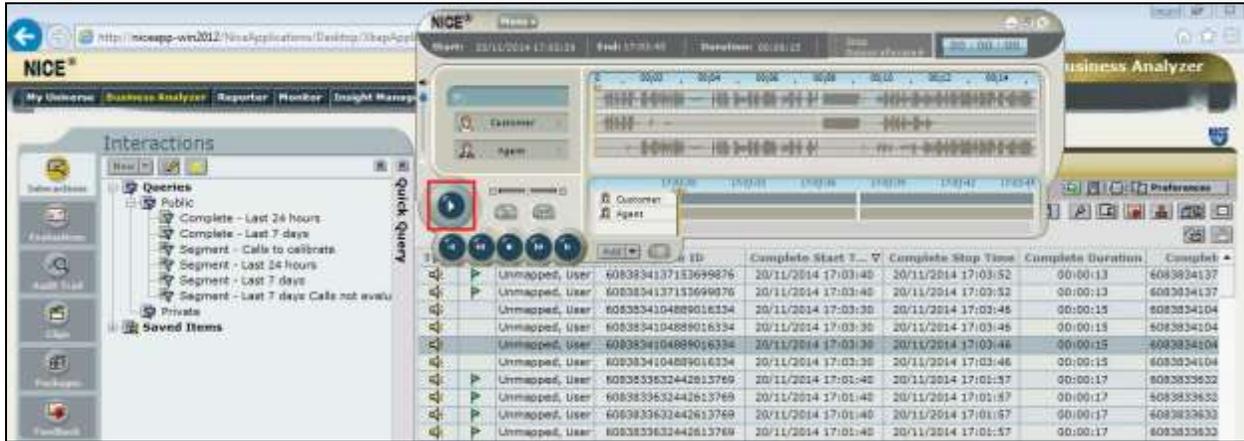
Click on **Business Analyser** at the top of the screen. Select **Interactions** from the left window and then navigate to **Queries** → **Public**.



Click on **Complete – Last 24 hours**. This should reveal all the recordings that took place over the previous 24 hours. Select the required recording from the list and double-click on this to play the recording.

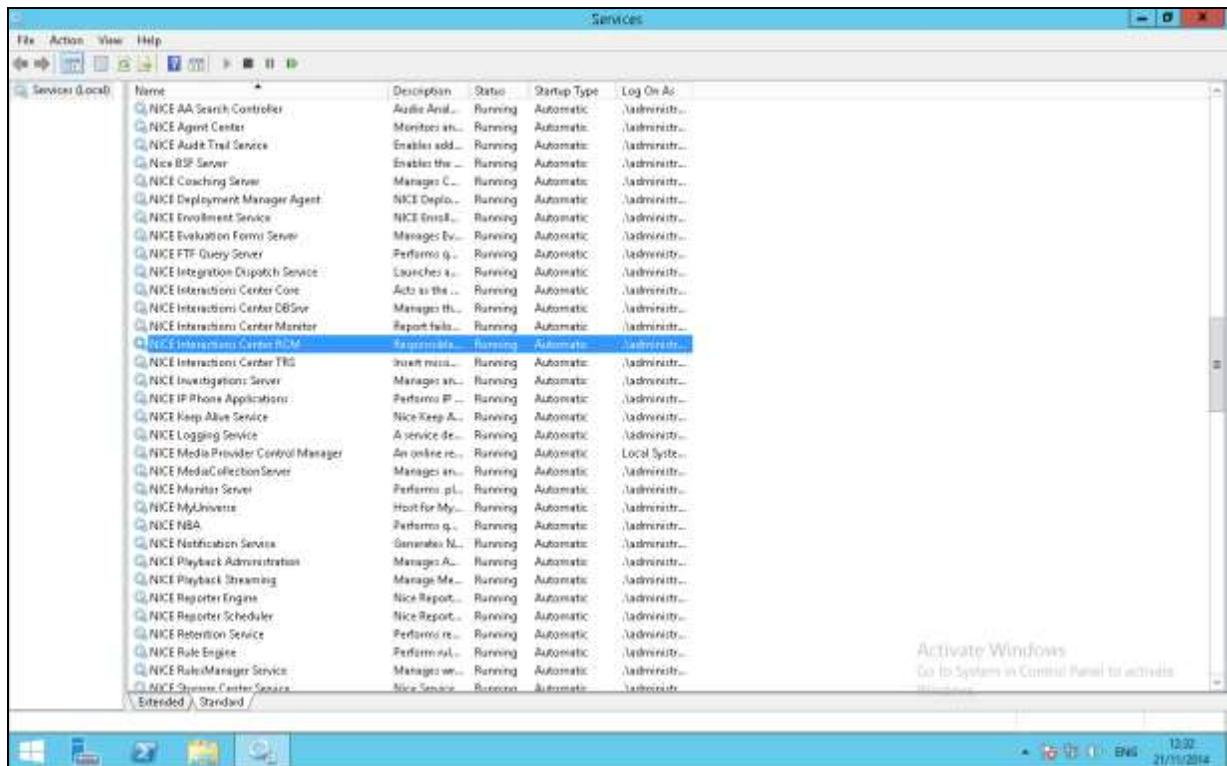


The NICE player is opened and the recording is presented for playback. Click on the **Play** icon highlighted below to play back the recording.



## 8.4. Verify NICE Services

If these recordings are not present or cannot be played back the NICE services may not be running or may need to be restarted. There are two separate servers as a part of this NICE Engage Platform. The NICE Application Server and the NICE Advanced Interactions Server can be logged into and checked to ensure all services beginning with NICE are running correctly. As a last resort both servers may need a reboot after the initial configuration.



## 9. Conclusion

These Application Notes describe the configuration steps required for NICE Engage Platform to successfully interoperate with Avaya Aura® Communication Manager R6.3 using Avaya Aura® Application Enablement Services R6.3 to connect to using DMCC Service Observe and Single Step Conference to record calls. All feature functionality and serviceability test cases were completed successfully with some issues and observations noted in **Section 2.2**.

## 10. Additional References

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

Product documentation for Avaya products may be found at <http://support.avaya.com>.

- [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 6.3*
- [4] *Avaya Aura® Session Manager Overview*, Doc # 03603323 *Avaya Aura® Contact Centre SIP Commissioning*, Doc # NN44400-511, Release 6.3

Product documentation for NICE products may be found at: <http://www.nice.com/>

# Appendix

## Avaya one-X® Agent Softphone

This is a printout of the Avaya one-X® Agent softphone used during compliance testing.

```
display station 2100                                     Page 1 of 5
                                                         STATION
Extension: 2100                                         Lock Messages? n          BCC: 0
  Type: 9630                                           Security Code: *         TN: 1
  Port: S00031                                         Coverage Path 1:        COR: 1
  Name: one-X Agent1                                   Coverage Path 2:        COS: 1
                                                         Hunt-to Station:        Tests? y

STATION OPTIONS
  Location:                                           Time of Day Lock Table:
  Loss Group: 19                                     Personalized Ringing Pattern: 1
                                                         Message Lamp Ext: 2100
  Speakerphone: 2-way                                Mute Button Enabled? y
  Display Language: english                          Button Modules: 0
Survivable GK Node Name:
  Survivable COR: internal                            Media Complex Ext:
  Survivable Trunk Dest? y                            IP SoftPhone? y
                                                         IP Video Softphone? n
                                                         Short/Prefixed Registration Allowed: default
                                                         Customizable Labels? Y
```

```
display station 2100                                     Page 2 of 5
                                                         STATION
FEATURE OPTIONS
  LWC Reception: spe                                  Auto Select Any Idle Appearance? n
  LWC Activation? y                                  Coverage Msg Retrieval? y
  LWC Log External Calls? n                          Auto Answer: none
  CDR Privacy? n                                     Data Restriction? n
  Redirect Notification? y                            Idle Appearance Preference? n
  Per Button Ring Control? n                         Bridged Idle Line Preference? n
  Bridged Call Alerting? n                           Restrict Last Appearance? y
  Active Station Ringing: single
                                                         EMU Login Allowed? n
  H.320 Conversion? n                                Per Station CPN - Send Calling Number?
  Service Link Mode: as-needed                        EC500 State: enabled
  Multimedia Mode: enhanced                           Audible Message Waiting? n
  MWI Served User Type:                               Display Client Redirection? n
  AUDIX Name:                                         Select Last Used Appearance? n
                                                         Coverage After Forwarding? s
                                                         Multimedia Early Answer? n
Remote Softphone Emergency Calls: as-on-local Direct IP-IP Audio Connections? y
  Emergency Location Ext: 2100                       Always Use? n IP Audio Hairpinning? n
```

```

display station 2100                                     Page 3 of 5
                                     STATION
Conf/Trans on Primary Appearance? n
Bridged Appearance Origination Restriction? n

Call Appearance Display Format: disp-param-default
IP Phone Group ID:
Enhanced Callr-Info Display for 1-Line Phones? n

ENHANCED CALL FORWARDING
Forwarded Destination      Active
Unconditional For Internal Calls To: 1000      n
External Calls To: 1000      n
Busy For Internal Calls To:                    n
External Calls To:                            n
No Reply For Internal Calls To:                n
External Calls To:                            n

SAC/CF Override: n

```

```

display station 2100                                     Page 4 of 5
                                     STATION
SITE DATA
Room:                               Headset? n
Jack:                               Speaker? n
Cable:                             Mounting: d
Floor:                             Cord Length: 0
Building:                           Set Color:

ABBREVIATED DIALING
List1:                               List2:                               List3:

BUTTON ASSIGNMENTS
1: call-appr                        5: manual-in                        Grp:
2: call-appr                        6: after-call                       Grp:
3: call-appr                        7: aux-work                         RC:  Grp:
4: auto-in                          Grp:                                8:

voice-mail

```

## Avaya 9620 H.323 Deskphone

This is a printout of the Avaya 9620 H.323 Deskphone used during compliance testing.

```
display station 2000                                     Page 1 of 5
                                                         STATION
Extension: 2000                                         Lock Messages? n          BCC: 0
Type: 9620                                             Security Code: *          TN: 1
Port: S00000                                          Coverage Path 1: 2       COR: 1
Name: Paul 2000                                       Coverage Path 2:         COS: 1
                                                         Hunt-to Station:         Tests? y

STATION OPTIONS
  Location:
  Loss Group: 19
  Speakerphone: 2-way
  Display Language: english
  Survivable GK Node Name:
  Survivable COR: internal
  Survivable Trunk Dest? y

  Time of Day Lock Table:
  Personalized Ringing Pattern: 1
  Message Lamp Ext: 2000
  Mute Button Enabled? y

  Media Complex Ext:
  IP SoftPhone? n

  IP Video? n
  Short/Prefixed Registration Allowed: default

  Customizable Labels? y
```

```
display station 2000                                     Page 2 of 5
                                                         STATION
FEATURE OPTIONS
  LWC Reception: spe
  LWC Activation? y
  LWC Log External Calls? n
  CDR Privacy? n
  Redirect Notification? y
  Per Button Ring Control? n
  Bridged Call Alerting? n
  Active Station Ringing: single

  H.320 Conversion? n
  Service Link Mode: as-needed
  Multimedia Mode: enhanced
  MWI Served User Type:
  AUDIX Name:

  Auto Select Any Idle Appearance? n
  Coverage Msg Retrieval? y
  Auto Answer: none
  Data Restriction? n
  Idle Appearance Preference? n
  Bridged Idle Line Preference? n
  Restrict Last Appearance? y

  EMU Login Allowed? n
  Per Station CPN - Send Calling Number? y
  EC500 State: enabled
  Audible Message Waiting? n
  Display Client Redirection? n
  Select Last Used Appearance? n
  Coverage After Forwarding? s
  Multimedia Early Answer? n
  Direct IP-IP Audio Connections? y
  Always Use? n IP Audio Hairpinning? n

  Emergency Location Ext: 2000
```

```

display station 2000                                     Page 3 of 5
                STATION
    Conf/Trans on Primary Appearance? n
    Bridged Appearance Origination Restriction? n

    Call Appearance Display Format: inter-location
    IP Phone Group ID:
    Enhanced Callr-Info Display for 1-Line Phones? n

                ENHANCED CALL FORWARDING
                Forwarded Destination                Active
Unconditional For Internal Calls To: 4000                n
                External Calls To: 4000                n
    Busy For Internal Calls To: 4202                    n
                External Calls To: 4202                n
    No Reply For Internal Calls To: 2101                 n
                External Calls To: 2101                n

    SAC/CF Override: n

```

```

display station 2000                                     Page 4 of 5
                STATION
SITE DATA
    Room:                Headset? n
    Jack:                Speaker? n
    Cable:              Mounting: d
    Floor:              Cord Length: 0
    Building:          Set Color:

ABBREVIATED DIALING
    List1:                List2:                List3:

BUTTON ASSIGNMENTS
1: call-appr                4: manual-in                Grp:
2: call-appr                5: after-call                Grp:
3: auto-in                Grp:                6: aux-work                RC:                Grp:

    voice-mail

```

## Avaya Agent LoginID

This is a printout of one of the agents used during compliance testing.

```
display agent-loginID 4400                                Page 1 of 3
                                AGENT LOGINID
                                Login ID: 4400
                                Name: Paul
                                TN: 1
                                COR: 1
                                Coverage Path:
                                Security Code:
                                AAS? n
                                AUDIX? n
                                LWC Reception: spe
                                LWC Log External Calls? n
                                AUDIX Name for Messaging:
                                LoginID for ISDN/SIP Display? n
                                Password:
                                Password (enter again):
                                Auto Answer: station
                                MIA Across Skills: system
                                ACW Agent Considered Idle: system
                                Aux Work Reason Code Type: system
                                Logout Reason Code Type: system
                                Maximum time agent in ACW before logout (sec): system
                                Forced Agent Logout Time: :
```

```
display agent-loginID 4400                                Page 2 of 3
                                AGENT LOGINID
                                Direct Agent Skill:
                                Call Handling Preference: skill-level
                                Service Objective? n
                                Local Call Preference? n
                                SN    RL SL      SN    RL SL      SN    RL SL      SN    RL SL
1: 33    1
2: 34    1
3:
4:
5:
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```

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