



Avaya Solution & Interoperability Test Lab

Application Notes for NICE Engage Platform R7.1 to interoperate with Avaya Aura® Communication Manager R8.1 and Avaya Aura® Application Enablement Services R8.1 using DMCC Multiple Registration for Stereo Recording - 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 R8.1, and Avaya Aura® Session Manager R8.1, and Avaya Aura® Application Enablement Services R8.1.

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 R7.1 to interoperate with the Avaya solution consisting of an Avaya Aura® Communication Manager R8.1, an Avaya Aura® Session Manager R8.1, and Avaya Aura® Application Enablement Services R8.1. NICE Engage Platform uses Communication Manager's Multiple Registration feature via the Application Enablement Services (AES) Device, Media, and Call Control (DMCC) interface and the Telephony Services Application Programming Interface (TSAPI) to capture the audio and call details for call recording on various Communication Manager H.323, SIP and Digital endpoints, listed in **Section 4**.

Device Media Call Control (DMCC) allows 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.

NICE Engage Platform provides the ability to record multi-channel interactions across the organization for regulatory compliance and to utilize these interactions for multiple business applications in order to extract insights and gain value. The platform tightly integrates with the telephony environment via CTI, APIs and SIP and stores the metadata in a single recording platform to ensure regulatory adherence and standardized workforce optimization processes across multiple channels. It provides comprehensive search tools and media retrieval, as well as a wide variety of Real-Time capabilities for PCI compliance and advanced applications.

The NICE Engage Platform uses the Multiple Registration method to record the calls, using the TSAPI connection to monitor the events necessary to start and stop the recordings. The application uses the AE Services DMCC service to register itself as a recording device at the target extension. When the target extension joins a call, the application automatically receives the call's aggregated RTP media streams via the recording device and records the call, in stereo.

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 Multiple Registration. 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.

Avaya recommends our customers implement Avaya solutions using appropriate security and encryption capabilities enabled by our products. The testing referenced in these DevConnect Application Notes included the enablement of supported encryption capabilities in the Avaya products. Readers should consult the appropriate Avaya product documentation for further information regarding security and encryption capabilities supported by those Avaya products.

Support for these security and encryption capabilities in any non-Avaya solution component is the responsibility of each individual vendor. Readers should consult the appropriate vendor-supplied product documentation for more information regarding those products.

For the testing associated with these Application Notes, the interface between Avaya systems and NICE Recording did not include use of any specific encryption features as requested by NICE.

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/Outbound calls** – Test call recording for inbound and outbound calls to the Communication Manager to and from PSTN callers.
- **Hold/Transferred/Conference calls** – Test call recording for calls transferred to and in conference with PSTN callers.
- **Feature calls** - Test call recording for calls that are parked or picked up using Call Park, Call Pickup, Bridged Appearance and Service Observing.
- **Calls to Elite Agents** – Test call recording for calls to Communication Manager agents logged into Avaya Agent for Desktop.
- **Serviceability testing** - The behavior of NICE Engage Platform under different simulated failure conditions.

2.2. Test Results

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

- When a call is Parked and Unparked, the first leg is recorded, the second leg (unparked call) is recorded but there is no RTP present when a SIP phone is unparking the call. Avaya is investigating the issue.
- Observing a station/user/extension that is not monitored from a station/user/extension that is monitored can cause no CTI events on the observer. Recordings will appear in NICE Business Analyser (NBA), according to pre-configured Total Recording Solution (TRS) insertion time out (default 5h). During testing, NICE decreased time out to get stored recordings shortly.

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 Multiple Registration to record calls. The NICE Application Server is setup for DMCC Multiple Registration and connects to the AES.

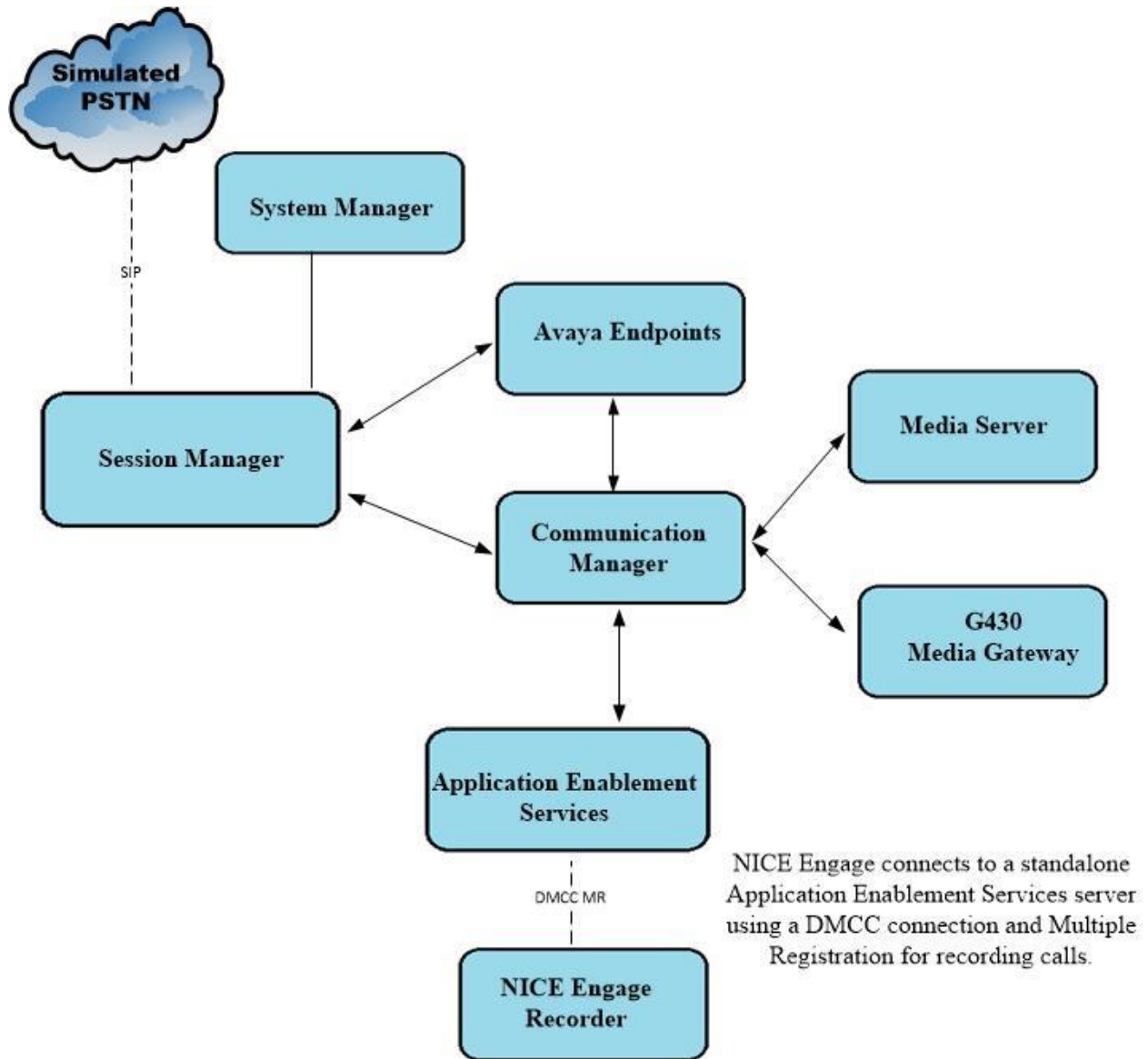


Figure 1: Connection of NICE Engage Platform R7.1 with Avaya Aura® Communication Manager R8.1, Avaya Aura® Session Manager R8.1 and Avaya Aura® Application Enablement Services R8.1

4. Equipment and Software Validated

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

| Avaya Equipment | Release/Version |
|--|---|
| Avaya Aura® System Manager running on a virtual server | 8.1.3.2 Build No. – 8.1.0.0.733078 Software Update Revision No: 8.1.3.2.1012646 Service Pack 2 |
| Avaya Aura® Session Manager running on a virtual server | 8.1.3.2 Build No. – 8.1.3.2.813207 |
| Avaya Aura® Communication Manager running on a virtual server | 8.1.3.2 – FP3SP2 R018x.01.0.890.0 Update ID 01.0.890.0-26989 |
| Avaya Aura® Application Enablement Services | 8.1.3.2 |
| Avaya Aura® Media Server | 8.0.2.184 |
| Avaya G430 Media Gateway | 41.16.0/1 |
| Avaya J179 H.323 Deskphone | 6.8502 |
| Avaya J189 SIP Deskphone | 4.0.10.1.2 |
| Avaya 9408 Digital Deskphone | V2.0 |
| Avaya Agent for Desktop | 2.0.6.8.3002 |
| NICE Equipment | Release/Version |
| NICE Engage Platform <ul style="list-style-type: none">- NICE Engage Application Server- NICE Engage AIR- NICE Engage NDM Server | 7.1 |

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 | Digital Loss Plan Modification? y | | |
| Async. Transfer Mode (ATM) Trunking? n | DS1 MSP? y | | |
| ATM WAN Spare Processor? n | DS1 Echo Cancellation? y | | |
| ATMS? 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 proc** and note the IP address for the **procr**.

| display node-names ip | | Page | 1 of 2 |
|-----------------------|--------------------|------|--------|
| IP NODE NAMES | | | |
| Name | IP Address | | |
| SM100 | 10.10.40.34 | | |
| aes8lvmpg | 10.10.40.16 | | |
| default | 0.0.0.0 | | |
| g450 | 10.10.40.15 | | |
| procr | 10.10.40.37 | | |

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 | Enabled | Local | Local | Remote | Remote | | |
| Type | | Node | Port | Node | 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 **aes81vmpg**.
- **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: | aes81vmpg | ***** | 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 | | | |
| | | COR: 1 | |
| Name: aes81vmpg | | | |

5.5. Configure H.323 Stations for Multi-Registration

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 Multi-Registration 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 during the NICE Recorder setup in **Section 7.1**. Note the **Security Code** and ensure that **IP SoftPhone** is set to y.

| | |
|----------------------------|--|
| change station 1001 | Page 1 of 6 |
| STATION | |
| Extension: 1001 | Lock Messages? n |
| Type: 9608 | Security Code: 1234 |
| Port: S00101 | Coverage Path 1: |
| Name: Extension | Coverage Path 2: |
| | Hunt-to Station: |
| STATION OPTIONS | |
| Loss Group: 19 | Time of Day Lock Table: |
| | Personalized Ringing Pattern: 1 |
| Speakerphone: 2-way | Message Lamp Ext: 1001 |
| Display Language: english | Mute Button Enabled? y |
| 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 |
| | IP Video Softphone? n |
| | Short/Prefixed Registration Allowed: default |

5.6. Configure SIP Stations for Multiple Registration

Each Avaya SIP endpoint or station that needs to be monitored for call recording will need to have “Type of 3PCC Enabled” is set to “Avaya” and “Softphone” set to “Yes”. Changes to SIP phones on Communication Manager must be carried out from System Manager. Access the System Manager using a Web Browser by entering **http://<FQDN>/network-login**, where **<FQDN>** is the fully qualified domain name of System Manager or the IP address of System Manager can be used as an alternative to the FQDN. Log in using appropriate credentials.

Note: The following shows changes a SIP extension and assumes that the SIP extension has been programmed correctly and is fully functioning.

Recommended access to System Manager is via FQDN.
[Go to central login for Single Sign-On](#)

If IP address access is your only option, then note that authentication will fail in the following cases:

- First time login with "admin" account
- Expired/Reset passwords

Use the "Change Password" hyperlink on this page to change the password manually, and then login.

Also note that single sign-on between servers in the same security domain is not supported when accessing via IP address.

This system is restricted solely to authorized users for legitimate business purposes only. The actual or attempted unauthorized access, use, or modification of this system is strictly prohibited.

Unauthorized users are subject to company disciplinary procedures and or criminal and civil penalties under state, federal, or other applicable domestic and foreign laws.

The use of this system may be monitored and recorded for administrative and security reasons. Anyone accessing this system expressly consents to such monitoring and recording, and is advised that if it reveals possible evidence of criminal activity, the evidence of such activity may be provided to law enforcement officials.

All users must comply with all corporate instructions regarding the protection of information assets.

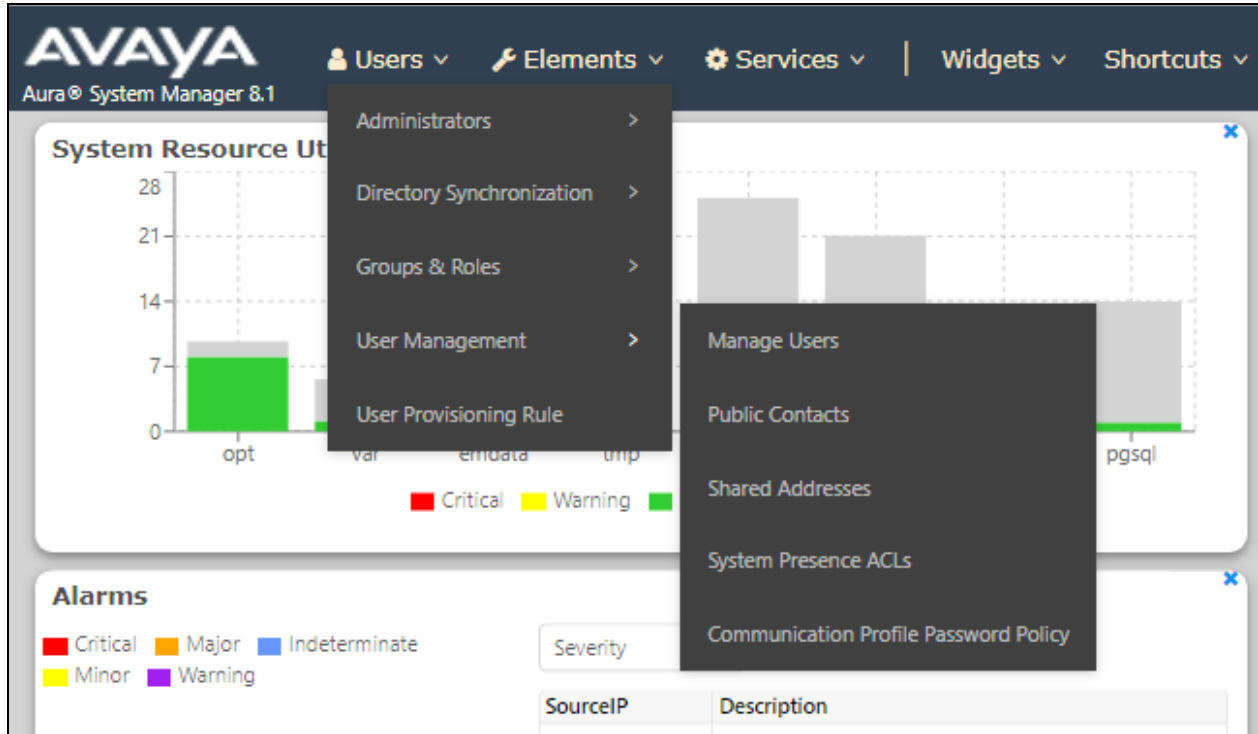
User ID:

Password:

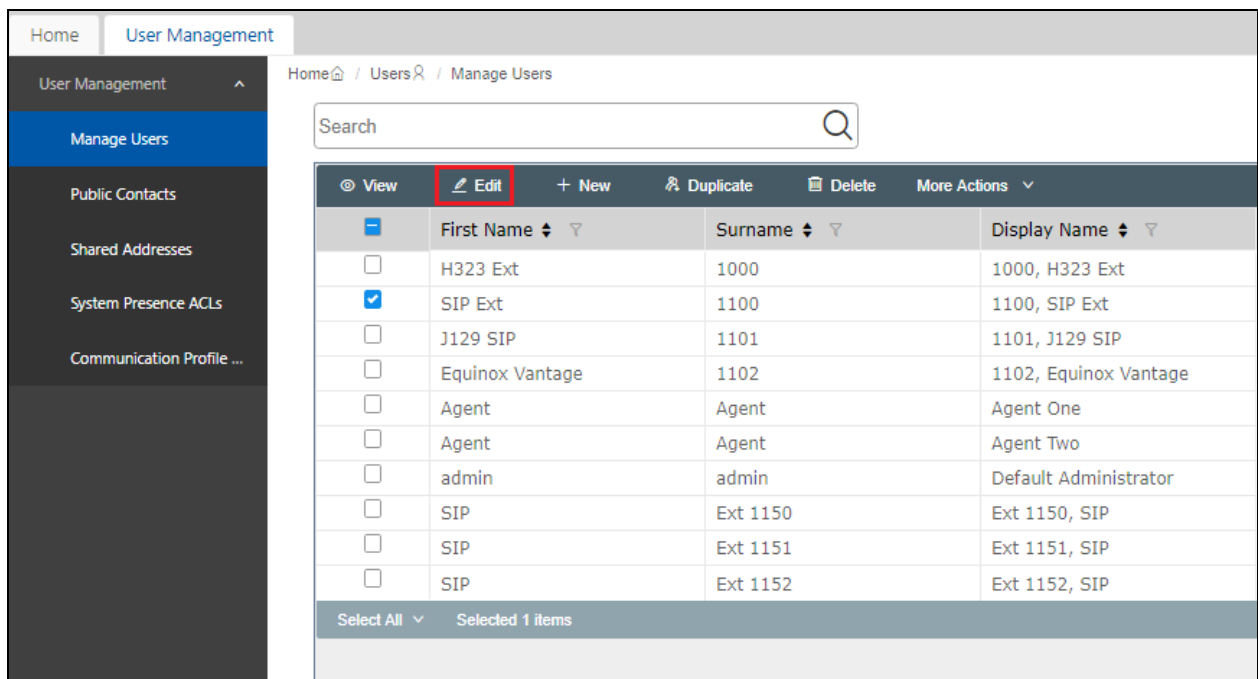
[Change Password](#)

Supported Browsers: Internet Explorer 11.x or Firefox 65.0, 66.0 and 67.0.

From the home page, click on **Users** → **User Management** → **Manage Users**, as shown below.



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



Click on the **CM Endpoint Profile** tab in the left window. Click on **Endpoint Editor** to make changes to the SIP station.

User Profile | Edit | 1100@devconnect.local

Commit & Continue Commit Cancel

Identity Communication Profile Membership Contacts

Communication Profile Password

PROFILE SET : Primary

Communication Address

PROFILES

Session Manager Profile ☒

Avaya Breeze® Profile ☐

CM Endpoint Profile ☒

* System : cm\$1xvmpg

* Profile Type : Endpoint

Use Existing Endpoints : ☐

* Extension : 1100

Template : Start typing...

* Set Type : 9641SIPCC

Security Code : Enter Security Code

Port : S000002

Voice Mail Number : 6666

Preferred Handle : Select

Calculate Route Pattern : ☐

Sip Trunk : aar

SIP URI : Select

Enhanced Callr-Info Display for 1-line phones : ☐

Delete on Unassign from User or on Delete User : ☒

Override Endpoint Name and Localized Name : ☒

Allow H.323 and SIP Endpoint Dual Registration : ☐

In the **General Options** tab ensure that **Type of 3PCC Enabled** is set to **Avaya** as is shown below. Click on **Done**, at the bottom of the screen, once this is set, (not shown).

General Options (G) * Feature Options (F) Site Data (S) Abbreviated Call Dialing (A)

Enhanced Call Fwd (E) Button Assignment (B) Profile Settings (P) Group Membership (M)

* Class of Restriction (COR) 1

* Emergency Location Ext 1100

* Tenant Number 1

* SIP Trunk aar

Coverage Path 1

Lock Message ☐

Multibyte Language Not Applicable

* Class Of Service (COS) 1

* Message Lamp Ext. 1100

Type of 3PCC Enabled Avaya

Coverage Path 2

Localized Display Name 1100, SIP Ext

Enable Reachability for Station Domain Control system

SIP URI

Primary Session Manager

IPv4: 10.10.40.32 IPv6:

Secondary Session Manager

Click on the **Feature Options** tab and ensure that **IP Softphone** is ticked as shown. Click on **Done** at the bottom of the screen once this is set.

The screenshot shows the 'Feature Options' tab selected. The 'IP SoftPhone' checkbox is checked and highlighted with a red box. Other settings include 'Active Station Ringing' set to 'single', 'MWI Served User Type' set to 'sip-adjunct', 'Per Station CPN - Send Calling Number' set to 'None', 'IP Phone Group ID' (empty), 'Remote Soft Phone Emergency Calls' set to 'as-on-local', 'LWC Reception' set to 'spe', 'AUDIX Name' (empty), 'Short/Prefixed Registration Allowed' set to 'default', 'Voice Mail Number' (empty), 'Auto Answer' set to 'none', 'Coverage After Forwarding' set to 'system', 'Display Language' set to 'english', 'Hunt-to Station' (empty), 'Loss Group' set to '19', 'Survivable COR' set to 'internal', 'Time of Day Lock Table' set to 'None', and 'Music Source' (empty). The 'Features' section includes checkboxes for 'Always Use', 'IP Audio Hairpinning', 'Bridged Call Alerting', 'Bridged Idle Line Preference', 'Coverage Message Retrieval' (checked), 'Data Restriction', 'Survivable Trunk Dest' (checked), 'Bridged Appearance Origination Restriction', 'Restrict Last Appearance' (checked), 'Idle Appearance Preference', 'IP SoftPhone' (checked), 'LWC Activation', 'CDR Privacy', 'Direct IP-IP Audio Connections' (checked), 'H.320 Conversion', 'IP Video Softphone', and 'Per Button Ring Control'.

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

The screenshot shows the 'User Profile | Edit | 1100@devconnect.local' interface. The 'Commit' button is highlighted with a red box. The 'CM Endpoint Profile' is selected under 'PROFILES'. The 'System' is set to 'cm\$txvmpg', 'Profile Type' is 'Endpoint', 'Extension' is '1100', 'Set Type' is '9641SIPCC', and 'Port' is 'S000002'. Other settings include 'Use Existing Endpoints' (unchecked), 'Template' (empty), 'Security Code' (empty), 'Voice Mail Number' (6666), 'Calculate Route Pattern' (unchecked), 'SIP URI' (Select), 'Enhanced Callr-Info Display for 1-line phones' (unchecked), 'Delete on Unassign from User or on Delete User' (checked), 'Allow H.323 and SIP Endpoint Dual Registration' (unchecked), 'Override Endpoint Name and Localized Name' (checked), and 'Preferred Handle' (Select). The 'Sip Trunk' is set to 'aar'.

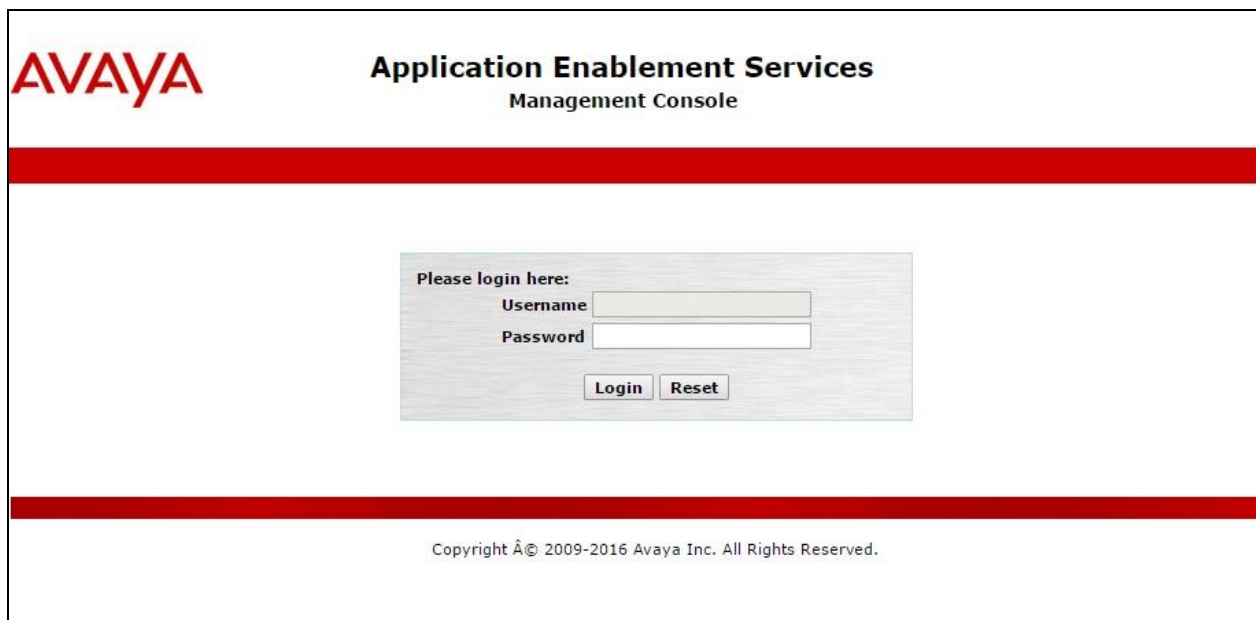
6. Configure Avaya Aura® Application Enablement Services

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
- Configure Networking Ports
- Create CTI User
- Configure Security Database

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 interface. At the top left is the Avaya logo. To its right, the text "Application Enablement Services" is displayed in a large, bold font, with "Management Console" in a smaller font below it. A thick red horizontal bar spans the width of the page. Below this bar is a light gray rectangular box containing the login form. The form has the text "Please login here:" followed by two input fields: "Username" and "Password". Below these fields are two buttons: "Login" and "Reset". Another thick red horizontal bar is located below the login box. At the bottom of the page, centered, is the copyright notice: "Copyright © 2009-2016 Avaya Inc. All Rights Reserved."

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

The screenshot shows the 'AE Services' management console. On the left is a navigation menu with options like CVLAN, DLG, DMCC, SMS, TSAPI, TWS, Communication Manager Interface, High Availability, Licensing, Maintenance, Networking, Security, Status, User Management, Utilities, and Help. The 'Licensing' option is highlighted. The main content area is titled 'AE Services' and contains an important note about restarting services. Below this is a table listing various services and their current status and license mode.

| Service | Status | State | License Mode | Cause* |
|-------------------------|----------------|---------|--------------|--------|
| ASAI Link Manager | N/A | Running | N/A | N/A |
| CVLAN Service | OFFLINE | Running | N/A | N/A |
| DLG Service | OFFLINE | Running | N/A | N/A |
| DMCC 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 |
| AE Services HA | Not Configured | N/A | N/A | N/A |

Below the table, there is a link to 'Status and Control' and a note about the cause column. At the bottom, there is a 'License Information' section stating that the user is licensed to run Application Enablement (CTI) release 8.x.

The TSAPI and DMCC licenses are user licenses issued by the Web License Manager to which the Application Enablement Services server is pointed to. From the left window open **Licensing** and click on **WebLM Server Access** as shown below.

The screenshot shows the 'Licensing' management console. The left navigation menu is expanded, showing options like AE Services, Communication Manager Interface, High Availability, Licensing, WebLM Server Address, WebLM Server Access, Reserved Licenses, Maintenance, Networking, Security, Status, User Management, Utilities, and Help. The 'WebLM Server Access' option is highlighted. The main content area is titled 'Licensing' and contains instructions on how to set up and maintain the WebLM, including a list of required information for each step. A red note at the bottom advises users to disable their pop-up blocker.

Licensing

If you are setting up and maintaining the WebLM, you need to use the following:

- WebLM Server Address

If you are importing, setting up and maintaining the license, you need to use the following:

- WebLM Server Access

If you want to administer TSAPI Reserved Licenses or DMCC Reserved Licenses, you need to use the following:

- Reserved Licenses

NOTE: Please disable your pop-up blocker if you are having difficulty with opening this page

The following screen shows the available licenses for **TSAPI** and **DMCC** users.

▼ Application_Enablement

View license capacity

View peak usage

ASBCE

▶ Session_Border_Controller_E_AE

AVAYA_OCEANA

▶ Avaya_Oceana

CCTR

▶ ContactCenter

CE

▶ COLLABORATION_ENVIRONMENT

COLLABORATION_DESIGNER

▶ Collaboration_Designer

COLLABORATIVE_BROWSING_SNAP-IN

▶ Collaborative_Browsing_Snap_In


COMMUNICATION_MANAGER

▶ Call_Center

▶ Communication_Manager

License File Host IDs:

Licensed Features

10 Items  Show

All ▼

| Feature (License Keyword) | Expiration date | Licensed capacity |
|--|-----------------|-------------------|
| Unified CC API Desktop Edition VALUE_AES_AEC_UNIFIED_CC_DESKTOP | permanent | 44 |
| CVLAN ASAI VALUE_AES_CVLAN_ASAI | permanent | 44 |
| Device Media and Call Control VALUE_AES_DMCC_DMC | permanent | 44 |
| AES ADVANCED SMALL SWITCH VALUE_AES_AEC_SMALL_ADVANCED | permanent | 4 |
| DLG VALUE_AES_DLG | permanent | 44 |
| TSAPI Simultaneous Users VALUE_AES_TSAPI_USERS | permanent | 44 |
| AES ADVANCED LARGE SWITCH VALUE_AES_AEC_LARGE_ADVANCED | permanent | 4 |
| CVLAN Proprietary Links VALUE_AES_PROPRIETARY_LINKS | permanent | 44 |

6.2. Switch Connection to Avaya Aura® Communication Manager

Typically, the connection between the AES and Communication Manager is setup as part of the initial installation and would not usually be outlined in these Application Notes. Due to the nature of this particular setup with two connections from Communication Manager to two separate AES's the switch connection will be displayed on this section. From the AES Management Console navigate to **Communication Manager Interface → Switch Connections**, the connection to Communication Manager should be present as shown below but if one is not present one can be added by clicking on **Add Connection**.

Application Enablement Services

Management Console

Welcome: User cust

Last login: Thu May 13 15:41:17 2021 from 192.168.40.240

Number of prior failed login attempts: 0

HostName/IP: aes81xvmpg/10.10.40.38

Server Offer Type: VIRTUAL_APPLIANCE_ON_VMWARE

SW Version: 8.1.3.1.0.7-0

Server Date and Time: Thu Jun 10 10:04:56 IST 2021

HA Status: Not Configured

Communication Manager Interface | Switch Connections

Home | Help | Logout

AE Services

Communication Manager Interface

Switch Connections

Dial Plan

High Availability

Licensing

Maintenance

Networking

Security

Switch Connections

| Connection Name | Processor Ethernet | Msg Period | Number of Active Connections |
|--|--------------------|------------|------------------------------|
| <input type="radio"/> cm81large | Yes | 30 | 0 |
| <input checked="" type="radio"/> cm81xvmpg | Yes | 30 | 1 |

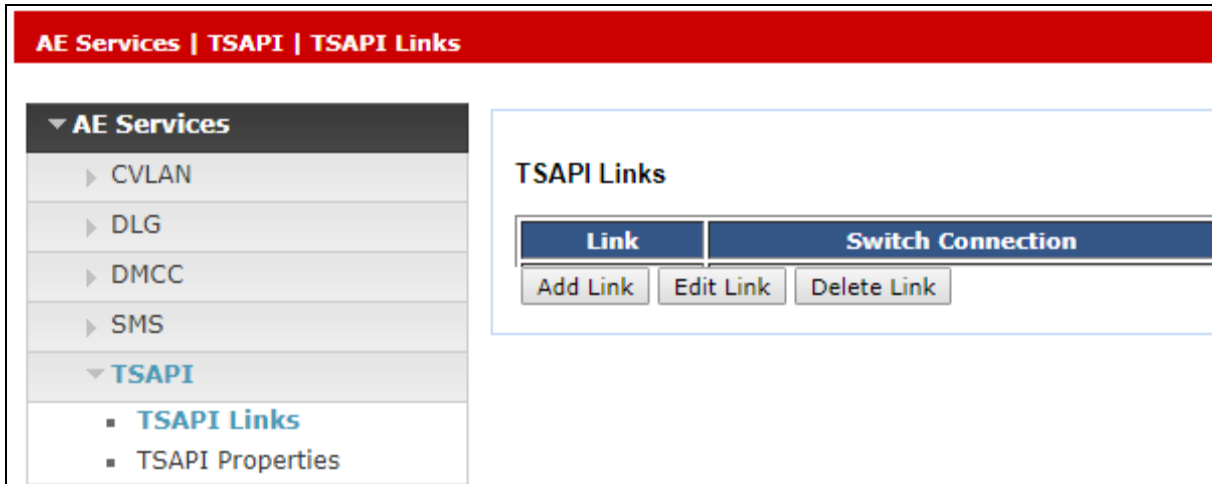
In the resulting screen, enter the **Switch Password**; the Switch Password must be the same as that entered into Communication Manager AE Services Administration screen via the **change ip-services** command, described in **Section 5.3**. A secure connection was established between the AES and Communication Manager, so the appropriate boxes were ticked, as shown below. 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.

| Name or IP Address | Status |
|--------------------|--------|
| 10.10.40.37 | In Use |

6.3. Administer TSAPI link

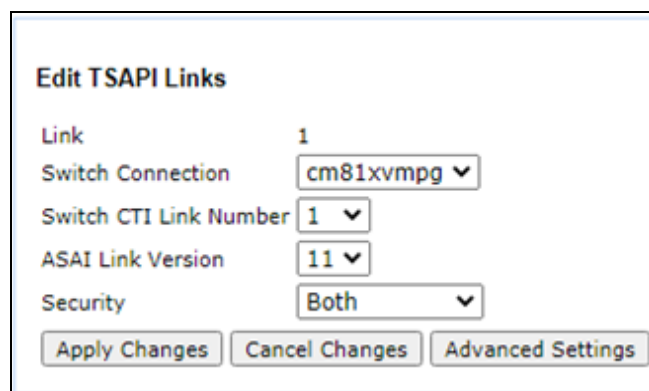
From the Application Enablement Services Management Console, select **AE Services** → **TSAPI** → **TSAPI Links**. Select **Add Link** button as shown in the screen below.



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 **cm81xvmpg**, 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:** **11** was used for compliance testing but the latest version available can be chosen).
- **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**.


Apply Changes to Link

Warning! Are you sure you want to apply the changes?
These changes can only take effect when the TSAPI server restarts.
 **Please use the Maintenance -> Service Controller page to restart the TSAPI server.**

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

| TSAPI Links | | | | |
|---|-------------------|-------------------|-------------------|----------|
| Link | Switch Connection | Switch CTI Link # | ASAI Link Version | Security |
| <input checked="" type="radio"/> 1 | cm81xvmpg | 1 | 8 | Both |
| <input type="button" value="Add Link"/> <input type="button" value="Edit Link"/> <input type="button" value="Delete Link"/> | | | | |

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**.



Application Enablement Services
Management Console

Maintenance | Service Controller

▶ AE Services

▶ Communication Manager Interface

▶ High Availability

▶ Licensing

▼ Maintenance

▶ Date Time/NTP Server

▶ Security Database

▶ Service Controller

▶ Server Data

▶ Networking

▶ Security

▶ Status

▶ User Management

▶ Utilities

▶ Help

Service Controller

| Service | Controller Status |
|---|-------------------|
| <input type="checkbox"/> ASAI Link Manager | Running |
| <input type="checkbox"/> DMCC Service | Running |
| <input type="checkbox"/> CVLAN Service | Running |
| <input type="checkbox"/> DLG Service | Running |
| <input type="checkbox"/> Transport Layer Service | Running |
| <input checked="" type="checkbox"/> TSAPI Service | Running |

For status on actual services, please use [Status and Control](#)

6.4. Identify Tlinks

Navigate to **Security** → **Security Database** → **Tlinks**. Verify the value of the **Tlink Name**. This will be needed to configure NICE Inform Recorder in **Section 7.1**.

The screenshot shows the NICE Inform Recorder web interface. At the top, a red header bar contains the text "Security | Security Database | Tlinks". On the left is a navigation menu with several categories: "AE Services", "Communication Manager Interface", "High Availability", "Licensing", "Maintenance", "Networking", "Security" (expanded), "Account Management", "Audit", "Certificate Management", "Enterprise Directory", "Host AA", "PAM", "Security Database" (expanded), and a list of sub-items under "Security Database": "Control", "CTI Users", "Devices", "Device Groups", "Tlinks" (highlighted in blue), "Tlink Groups", and "Worktops". The main content area on the right is titled "Tlinks". It contains a "Tlink Name" label, two radio button options, and a "Delete Tlink" button. The first radio button is selected and corresponds to the value "AVAYA#CM81XVMPG#CSTA#AES81XVMPG". The second radio button corresponds to "AVAYA#CM81XVMPG#CSTA-S#AES81XVMPG".

Security | Security Database | Tlinks

Tlinks

Tlink Name

☒ AVAYA#CM81XVMPG#CSTA#AES81XVMPG

☐ AVAYA#CM81XVMPG#CSTA-S#AES81XVMPG

Delete Tlink

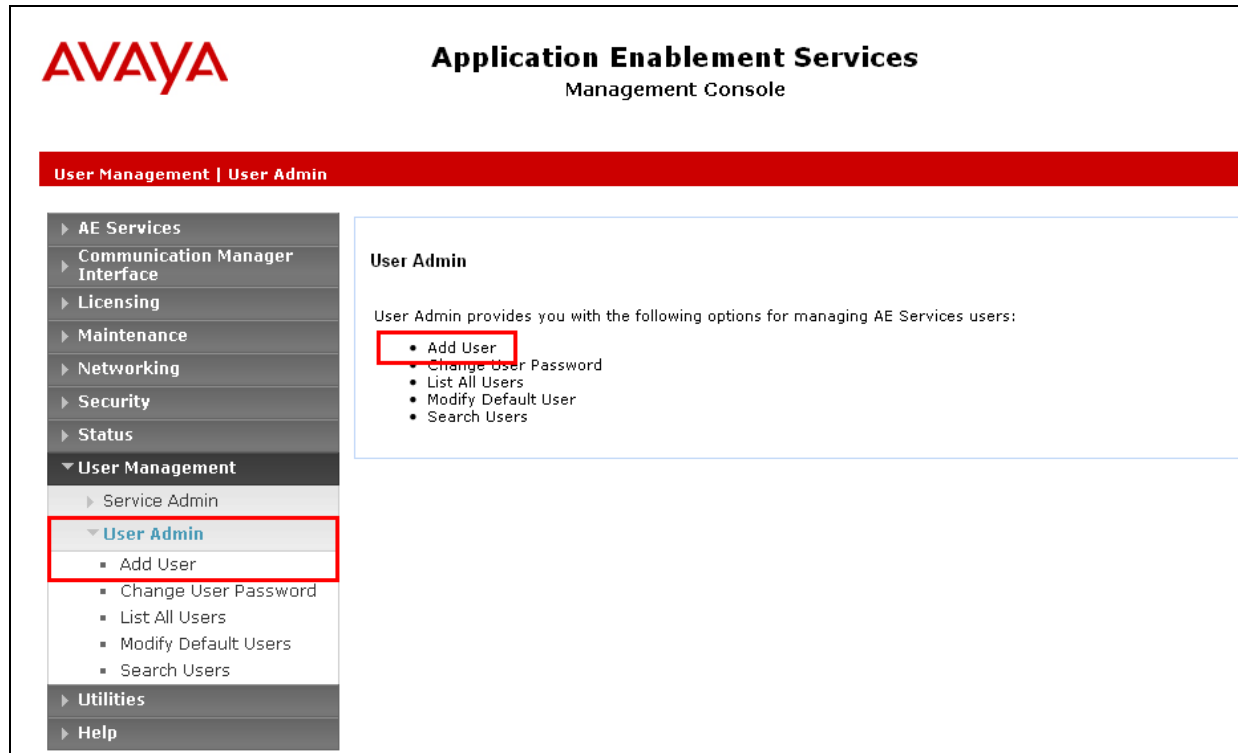
6.5. Enable TSAPI and DMCC Ports

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**.

| Networking Ports | | | | |
|--|------------------------------------|-----------------------------------|----------------------------------|-----------------------|
| <ul style="list-style-type: none"> ▶ AE Services ▶ Communication Manager Interface High Availability ▶ Licensing ▶ Maintenance ▼ Networking AE Service IP (Local IP) Network Configure Ports TCP/TLS Settings ▶ Security ▶ Status ▶ User Management ▶ Utilities ▶ Help | Ports | | | |
| | CVLAN Ports | | | Enabled Disabled |
| | Unencrypted TCP Port | 9999 | <input checked="" type="radio"/> | <input type="radio"/> |
| | Encrypted TCP Port | <input type="text" value="9998"/> | <input checked="" type="radio"/> | <input type="radio"/> |
| | <hr/> | | | |
| | DLG Port | TCP Port | 5678 | |
| | TSAPI Ports | | | Enabled Disabled |
| | TSAPI Service Port | 450 | <input checked="" type="radio"/> | <input type="radio"/> |
| | Local TLINK Ports | | | |
| | TCP Port Min | 1024 | | |
| TCP Port Max | 1039 | | | |
| Unencrypted TLINK Ports | | | | |
| TCP Port Min | <input type="text" value="1050"/> | | | |
| TCP Port Max | <input type="text" value="1065"/> | | | |
| Encrypted TLINK Ports | | | | |
| TCP Port Min | <input type="text" value="1066"/> | | | |
| TCP Port Max | <input type="text" value="1081"/> | | | |
| <hr/> | | | | |
| DMCC Server Ports | | | Enabled Disabled | |
| Unencrypted Port | <input type="text" value="4721"/> | <input checked="" type="radio"/> | <input type="radio"/> | |
| Encrypted Port | <input type="text" value="4722"/> | <input checked="" type="radio"/> | <input type="radio"/> | |
| TR/87 Port | <input type="text" value="4723"/> | <input checked="" type="radio"/> | <input type="radio"/> | |
| <hr/> | | | | |
| H.323 Ports | | | | |
| TCP Port Min | <input type="text" value="20000"/> | | | |
| TCP Port Max | <input type="text" value="29999"/> | | | |
| Local UDP Port Min | <input type="text" value="20000"/> | | | |
| Local UDP Port Max | <input type="text" value="29999"/> | | | |
| | | | Enabled Disabled | |
| Server Media | | <input checked="" type="radio"/> | <input type="radio"/> | |

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.



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.

AVAYA **Application Enablement Services**
Management Console

User Management | User Admin | Add User

Add User

Fields marked with * can not be empty.

| | |
|--------------------|-------|
| * User Id | NICE |
| * Common Name | NICE |
| * Surname | NICE |
| * User Password | ***** |
| * Confirm Password | ***** |
| Admin Note | |
| Avaya Role | None |
| Business Category | |
| Car License | |
| CM Home | |
| Css Home | |
| CT User | Yes |
| Department Number | |
| Display Name | |
| Employee Number | |
| Employee Type | |

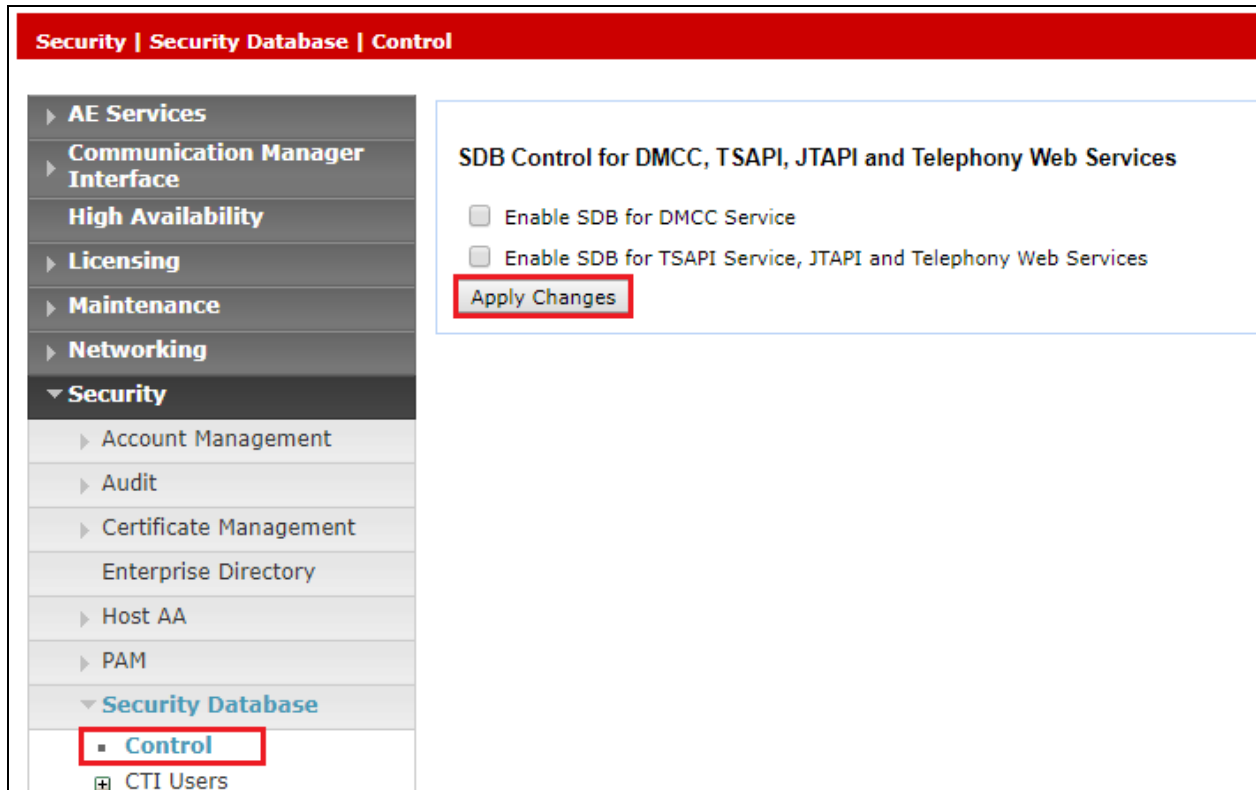
Scroll down and click on **Apply Changes** (not shown).

6.7. Configure Security Database

For compliance testing associated with these Application Notes the Security Database was not enabled and the user associated with NICE was given unrestricted access.

6.7.1. Disable the Security Database Control

Navigate to **Security** → **Security Database** → **Control** as shown below. Ensure that no boxes are ticked and click on **Apply Changes** if necessary.



6.7.2. Associate Devices with CTI User

Navigate to **Security** → **Security Database** → **CTI Users** → **List All Users**. Select the CTI user added in **Section 6.6** and click on **Edit Users**.

The screenshot shows the Avaya Application Enablement Services Management Console. The left sidebar contains a navigation menu with 'Security' expanded, and 'CTI Users' > 'List All Users' selected. The main area displays a table of CTI Users. The 'nice' user 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 |

Buttons: **Edit**, **List All**

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 'nice' user. The 'Unrestricted Access' checkbox is checked. The 'Apply Changes' button is highlighted.

Edit CTI User

User Profile:

| | |
|---------------------|-------------------------------------|
| 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 |
| Calls On A Device Monitoring | None |
| Call Monitoring | <input type="checkbox"/> |

Routing Control:

| | |
|---------------------------------|------|
| Allow Routing on Listed Devices | None |
|---------------------------------|------|

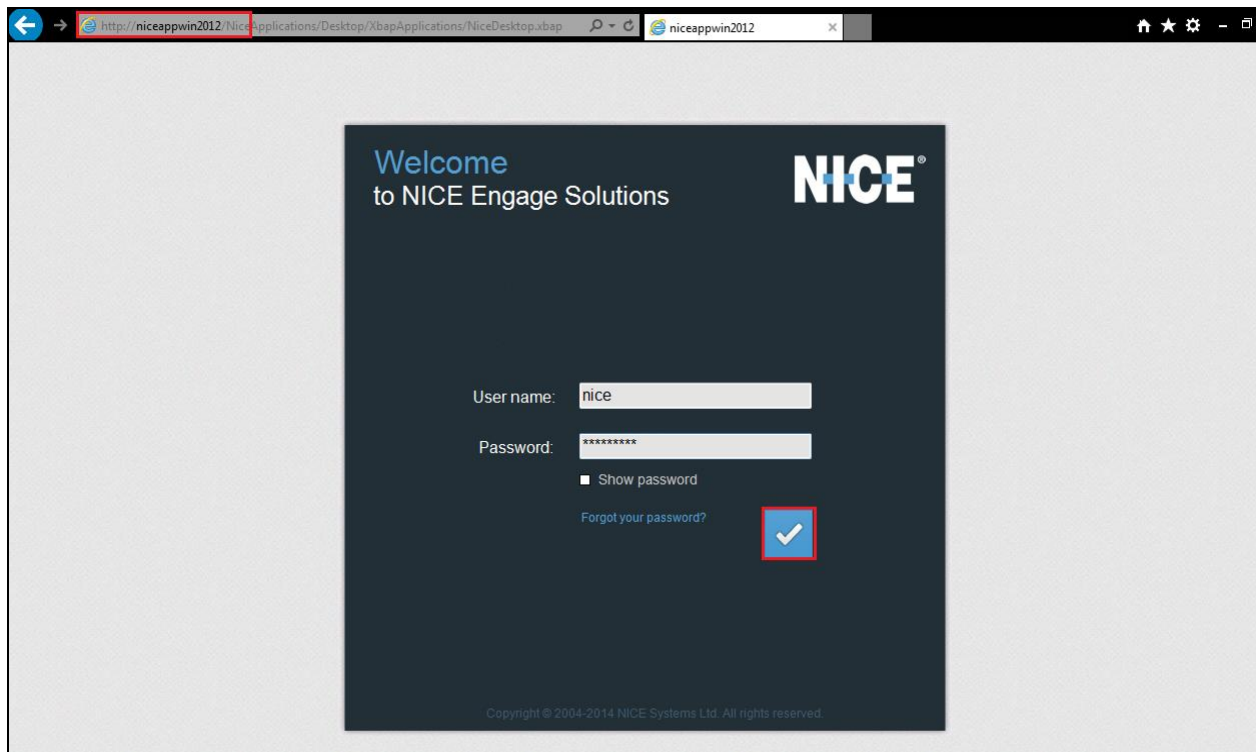
Buttons: **Apply Changes**, **Cancel Changes**

7. Configure NICE Engage Platform

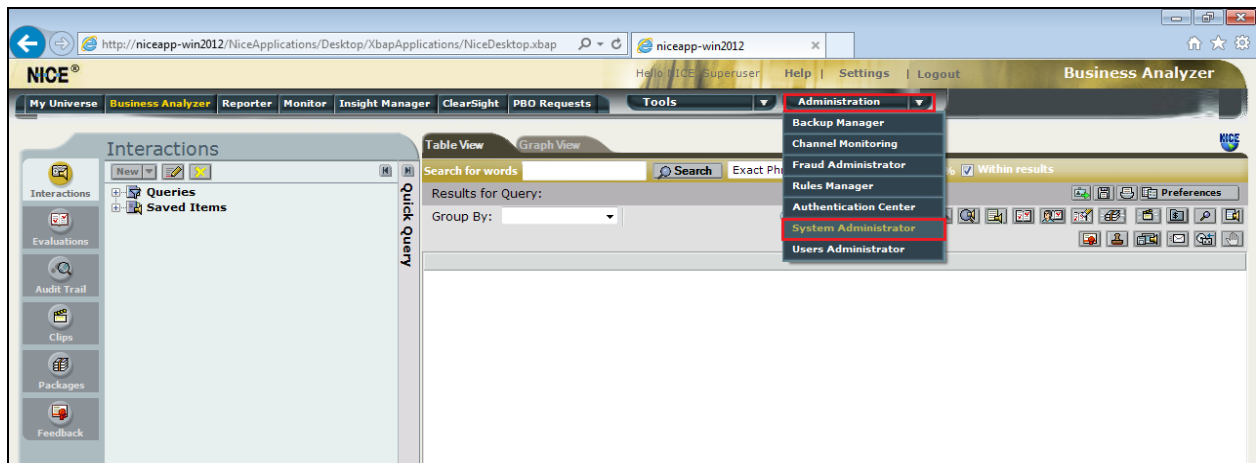
The installation of NICE Engage Platform is usually carried out by an engineer from NICE and is outside the scope of these Application Notes. For information on the installation of the NICE Engage Platform, contact NICE as per the information provided in **Section 2.3**.

The following sections will outline the process involved in connecting the NICE Engage Platform to the Avaya Solution.

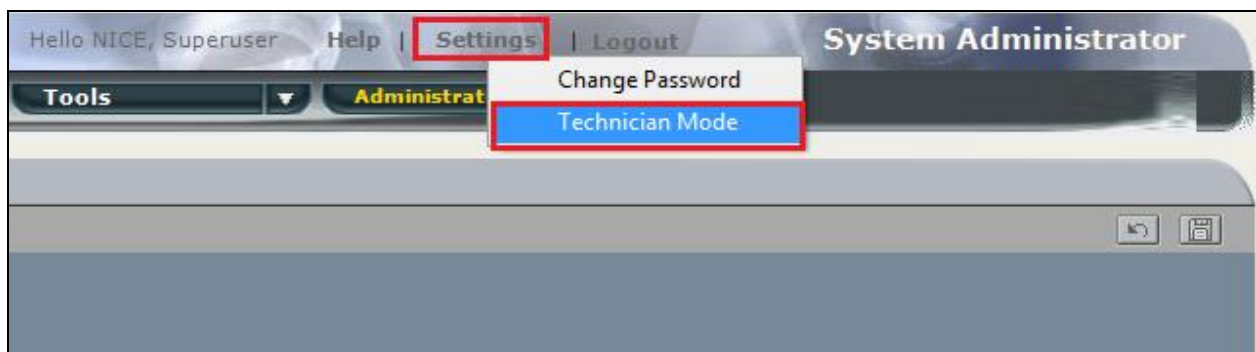
All configuration of the NICE Engage Platform for connection with the AES is performed using a web browser connecting to the NICE Engage Application Server. Open a web browser as shown navigate to **http://<NICEEngageApplicationServerIP>/Nice** as shown below and enter the proper credentials and click on **Login**.



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

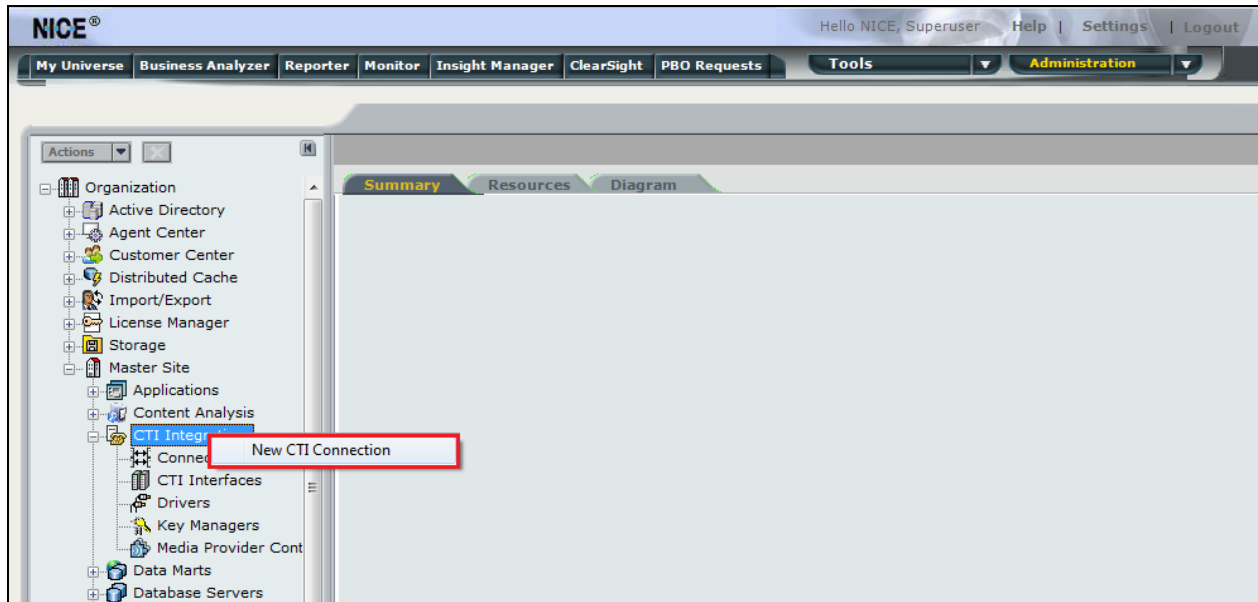


Before any changes can be made, switch to **Technician Mode** by clicking into **Settings** at the top of the screen as shown below.

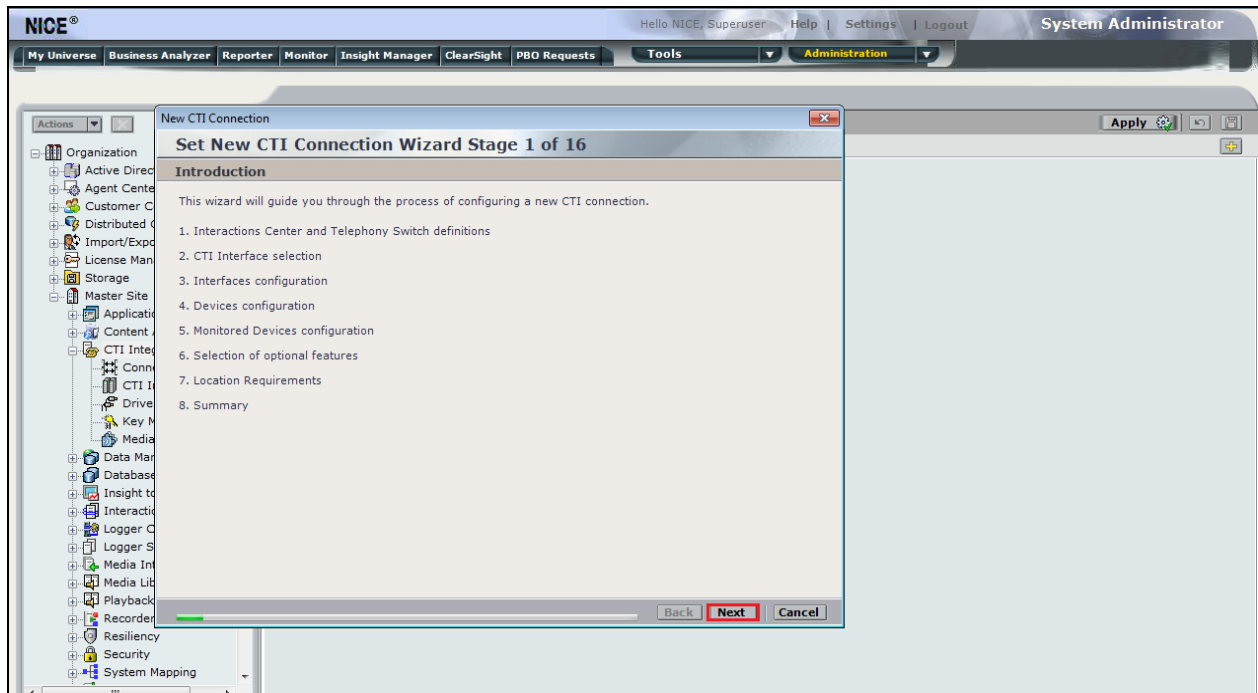


7.1. New CTI Connection

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 Multiple Registration type of call recording. Click on **Next** to continue.



The value for **Regular Interactions Center (IC)** 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.

New CTI Connection

Set New CTI Connection Wizard Stage 2 of 16

Interactions Center Switch

Attach CTI to Interactions Center Server:

☒ Regular Interactions Center: IC

☐ Interactions Center Cluster:

☐ Use existing Telephony Switch:

☒ Define new Telephony Switch:

Switch Type: Avaya CM

Switch Name: DevConnectCM

Advanced >>

Back Next Cancel

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

New CTI Connection

Set New CTI Connection Wizard Stage 3 of 16

Interface Type

CTI Interface Type

Avaya CM CTI Interface: AES TSAPI

Avaya Communication Manager
Avaya Application Enablement Services (AES) / Avaya CT - TSAPI

☐ VolP Mapping: AES SMS

☐ Additional VolP Mapping: Generic SIP Mapper

☒ Active Recording: DMCC (Advanced Interaction Recorder)

Avaya Communication Manager
Device Media and Call Control

Back Next Cancel

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

New CTI Connection

Set New CTI Connection Wizard Stage 4 of 16

Interface Parameters

CTI Interface Details

Interface Connection Details

Mandatory fields are marked in bold

| Parameter | Value |
|-----------------------|-------|
| ServerName | |
| LoginID | |
| Password | |
| UseWarmStandBy | No |

Description: Server connection name.

Additional Interface Parameters

Back Next Cancel

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

New CTI Connection

Set New CTI Connection Wizard Stage 4 of 16

Interface Parameters

CTI Interface Details

Interface Connection Details

Mandatory fields are marked in bold

| Parameter | Value |
|-----------------------|-------|
| ServerName | |
| LoginID | |
| Password | |
| UseWarmStandBy | No |

Description: Server connection name.

Additional Interface Parameters

Back Next Cancel

Set Parameter Value

Interface Connection Parameter

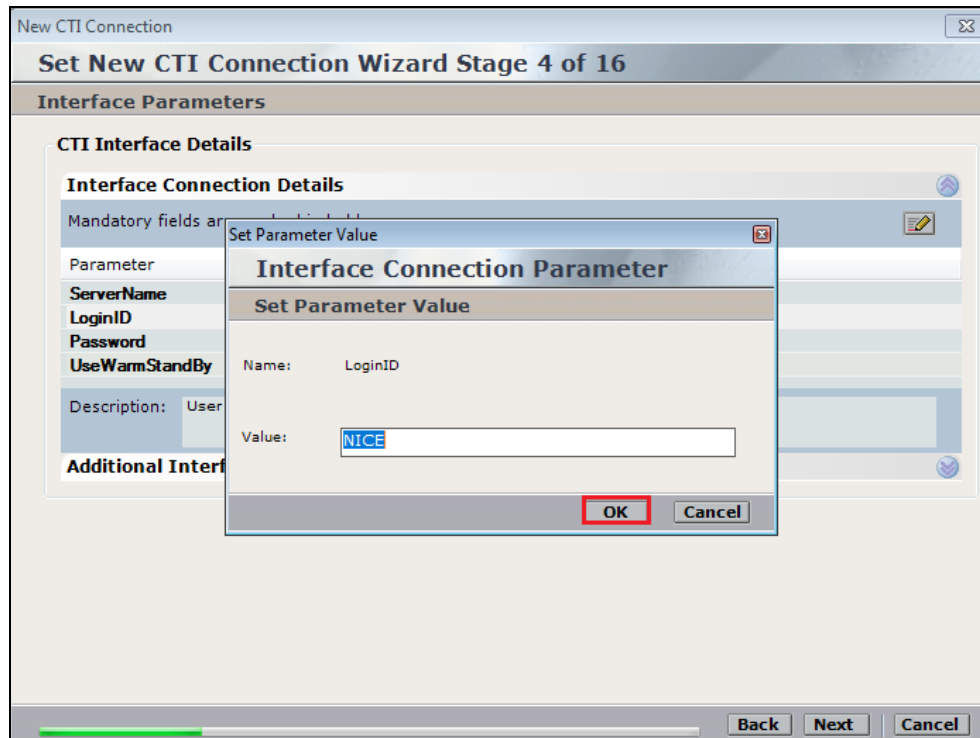
Set Parameter Value

Name: ServerName

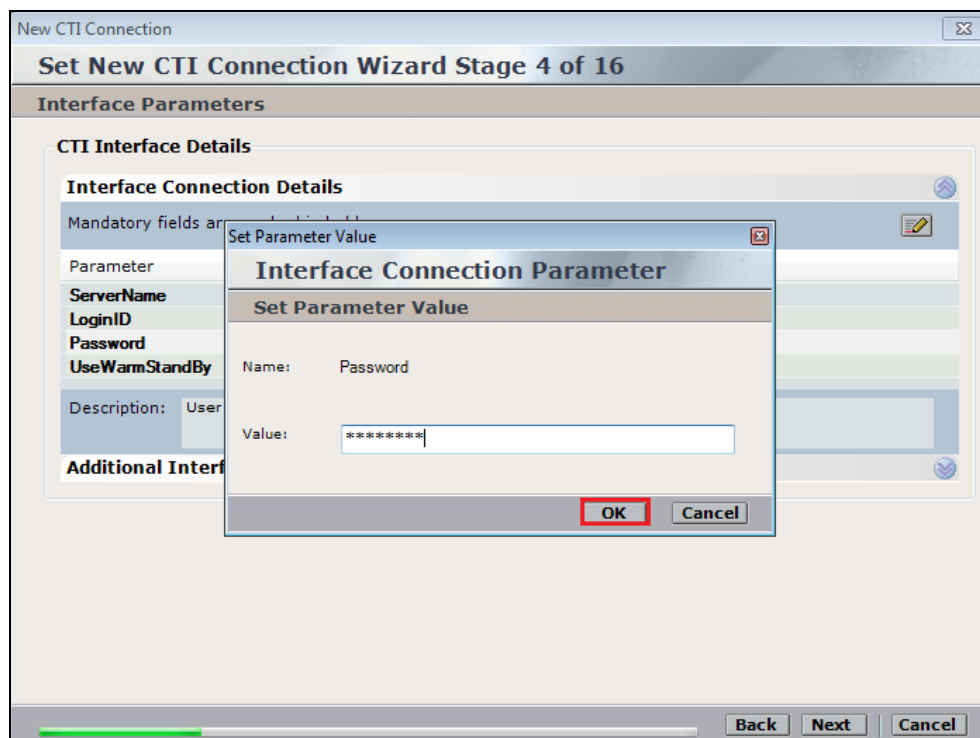
Value: AVAYA#CM81XVMPG#CSTA#AES81XVMPG

OK Cancel

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. Under 'CTI Interface Details', the 'Interface Connection Details' table is populated with the following values:

| Parameter | Value |
|-----------------------|---------------------------------|
| ServerName | AVAYA#CM81XVMPG#CSTA#AES81XVMPG |
| LoginID | nice |
| Password | ***** |
| UseWarmStandBy | No |

Below the table, the 'Description' field contains the text 'Is warm standby supported?'. The 'Additional Interface Parameters' section is currently empty. At the bottom of the wizard, the 'Next' button is highlighted with a red border.

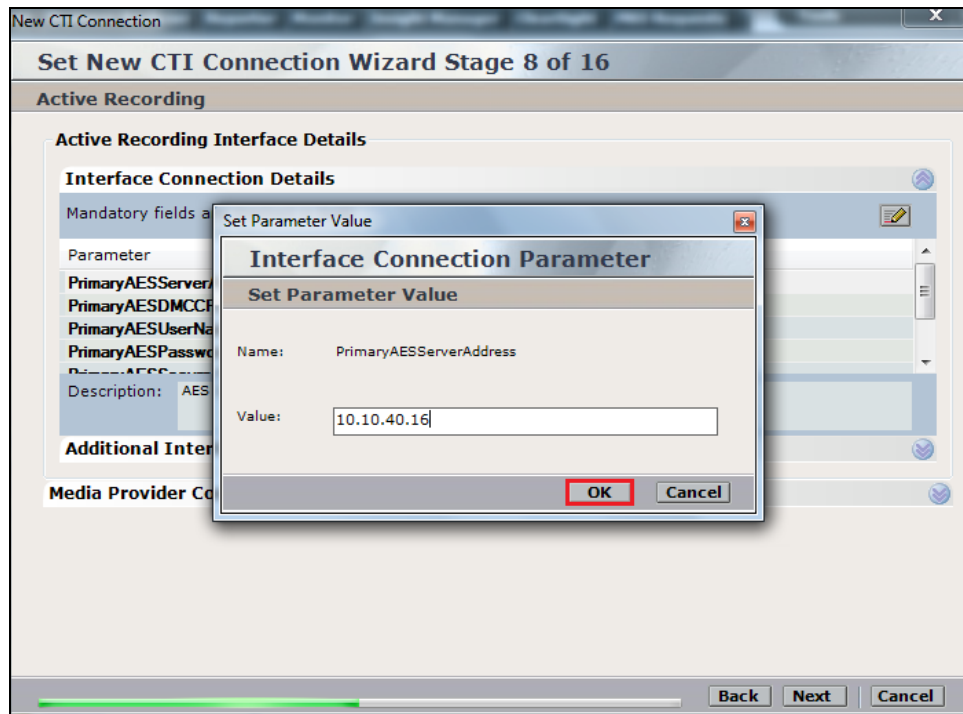
The values below must be filled in by double-clicking on each **Parameter**.

The screenshot shows the 'Set New CTI Connection Wizard Stage 8 of 16' window. The 'Active Recording' section is active. Under 'Active Recording Interface Details', the 'Interface Connection Details' table is populated with the following values:

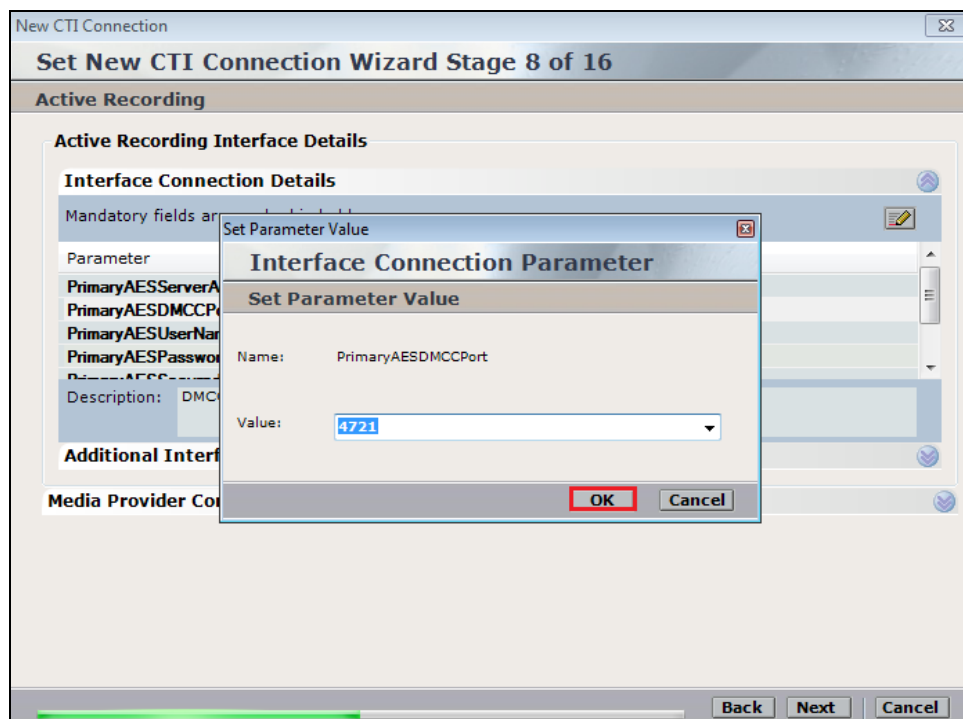
| Parameter | Value |
|------------------------------------|-------|
| PrimaryAESServerAddress | |
| PrimaryAESDMCCPort | 4722 |
| PrimaryAESUserName | |
| PrimaryAESPassword | |
| PrimaryAESStandByConnection | TRUE |

Below the table, the 'Description' field is empty. The 'Additional Interface Parameters' section is currently empty. At the bottom of the wizard, the 'Next' button is highlighted with a red border.

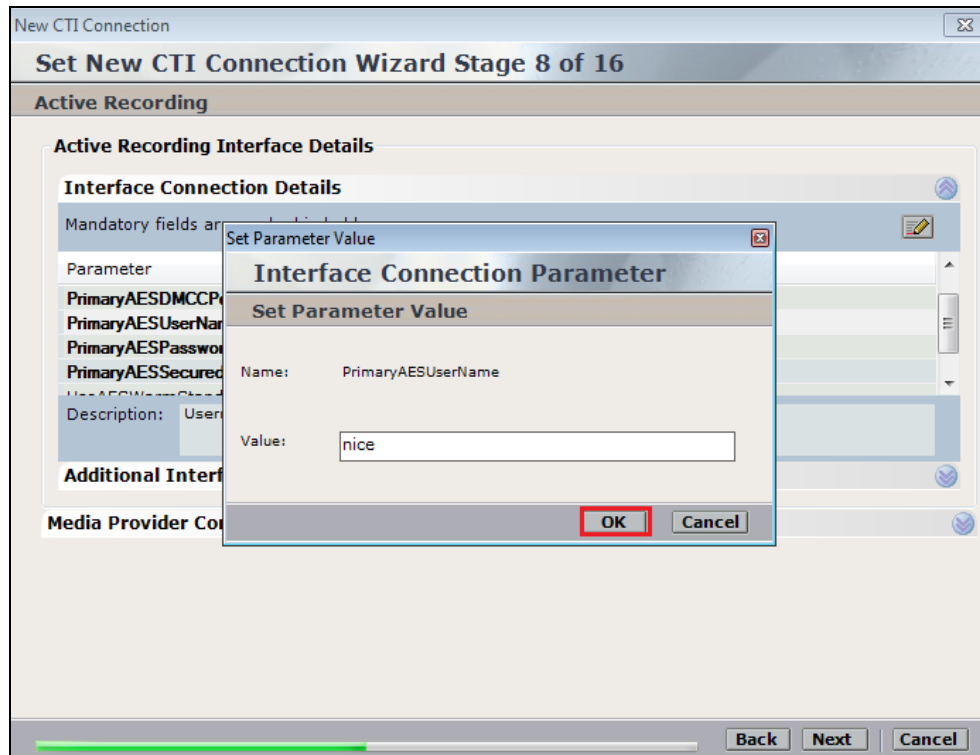
Enter the **Value** for the **AESServerAddress**, note this is the IP address of the AES server. Click on **OK**.



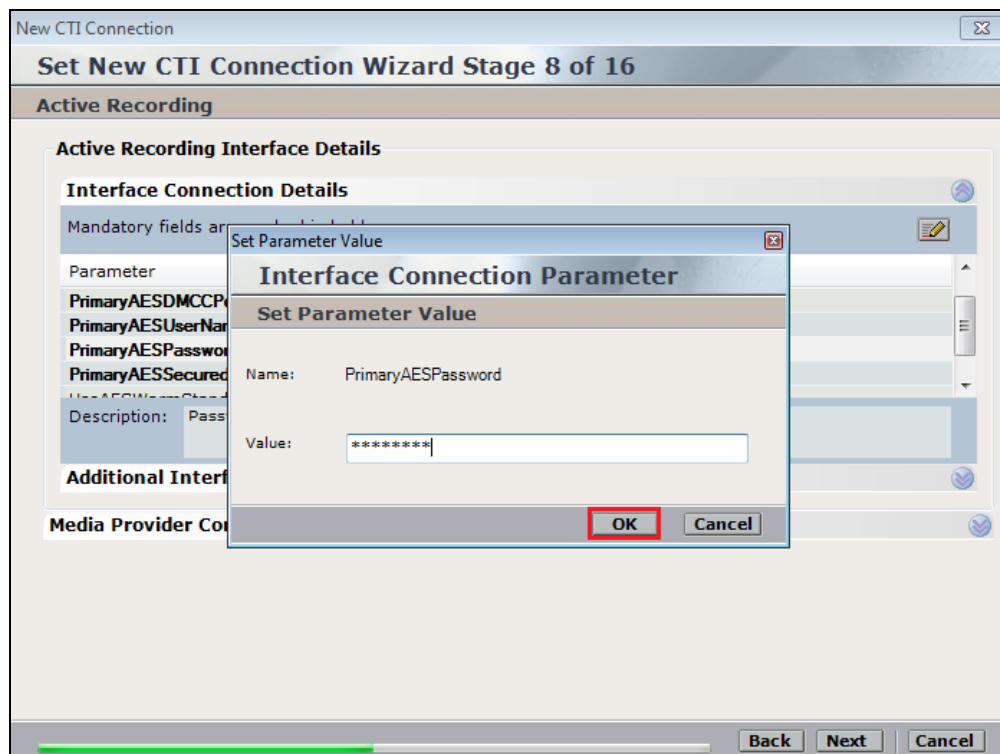
Enter the **Value** for the **AESDMCCPort**, note this will be the same port that was configured in **Section 6.5**. In this example the unencrypted port **4721** is entered.



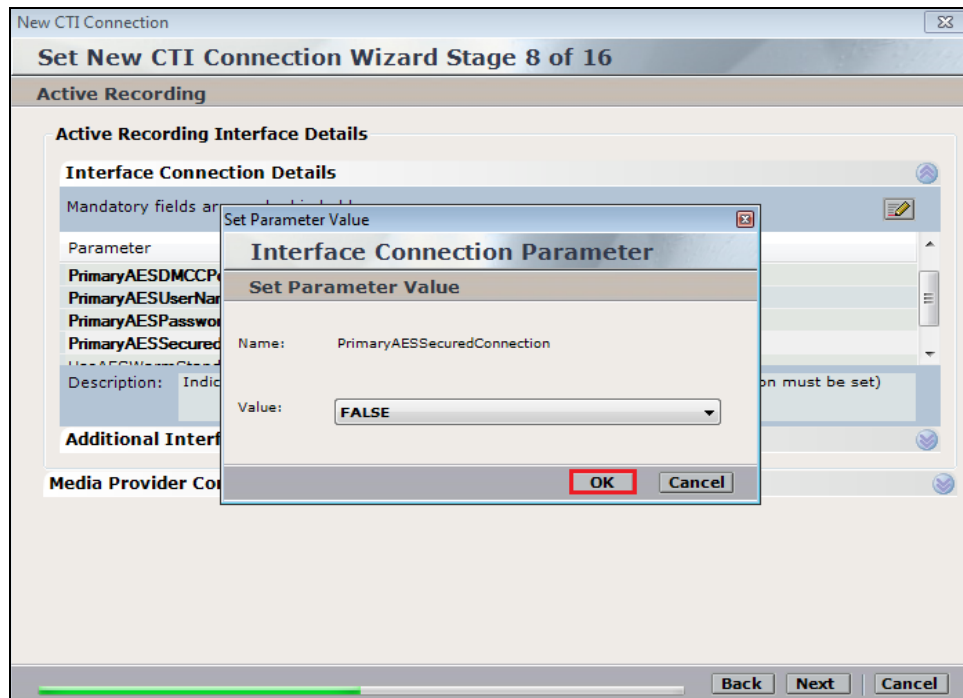
As before, enter the username that was created in **Section 6.6** and click on **OK**.



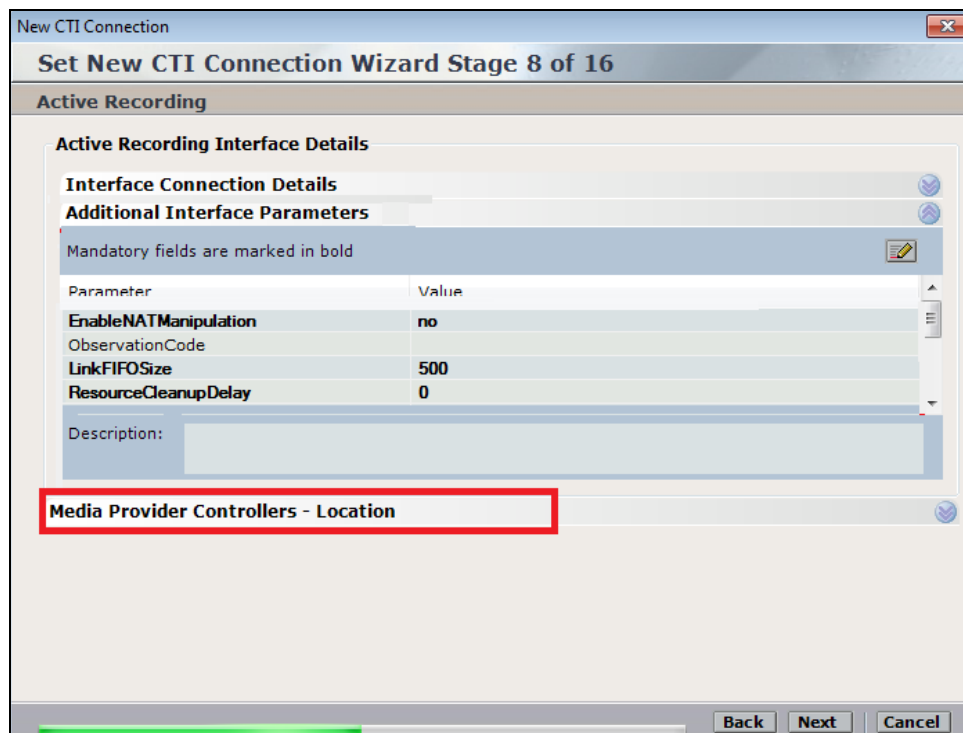
Enter the password that was created in **Section 6.6** and click on **OK**.



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

New CTI Connection

Set New CTI Connection Wizard Stage 8 of 16

Active Recording

Active Recording Interface Details

Interface Connection Details

Additional Interface Parameters

Media Provider Controllers - Location

Media Provider Location

Server IP/Hostname: NICEActive2012

Connection Manager Port: 62094

Media Provider Controllers:

| IP/Hostname | CM Port |
|-------------|---------|
| | |
| | |
| | |

Back Next Cancel

Click on **Next** to continue.

New CTI Connection

Set New CTI Connection Wizard Stage 8 of 16

Active Recording

Active Recording Interface Details

Interface Connection Details

Additional Interface Parameters

Media Provider Controllers - Location

Media Provider Location

Server IP/Hostname

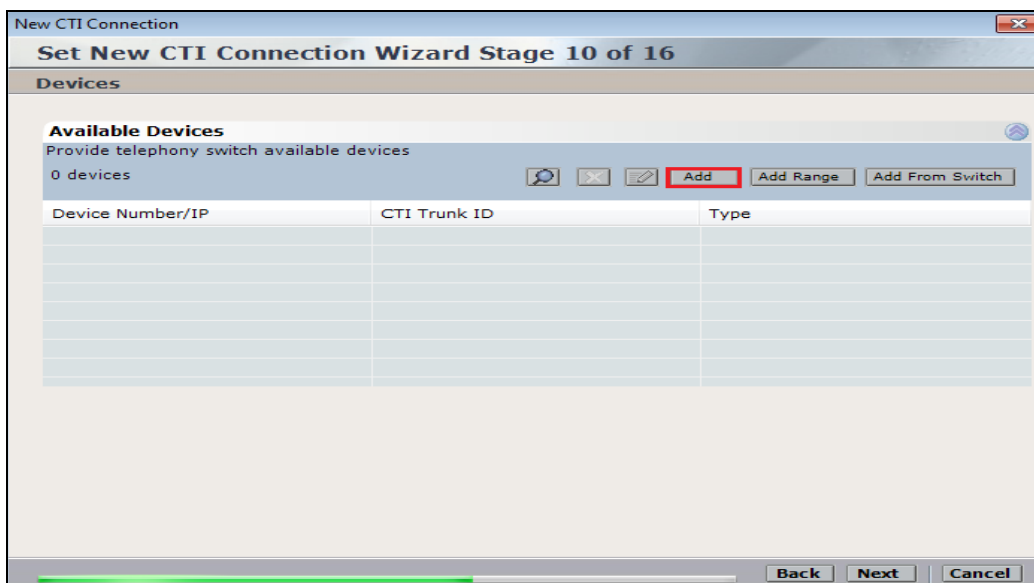
Connection Manager Port: 62094

Media Provider Controllers:

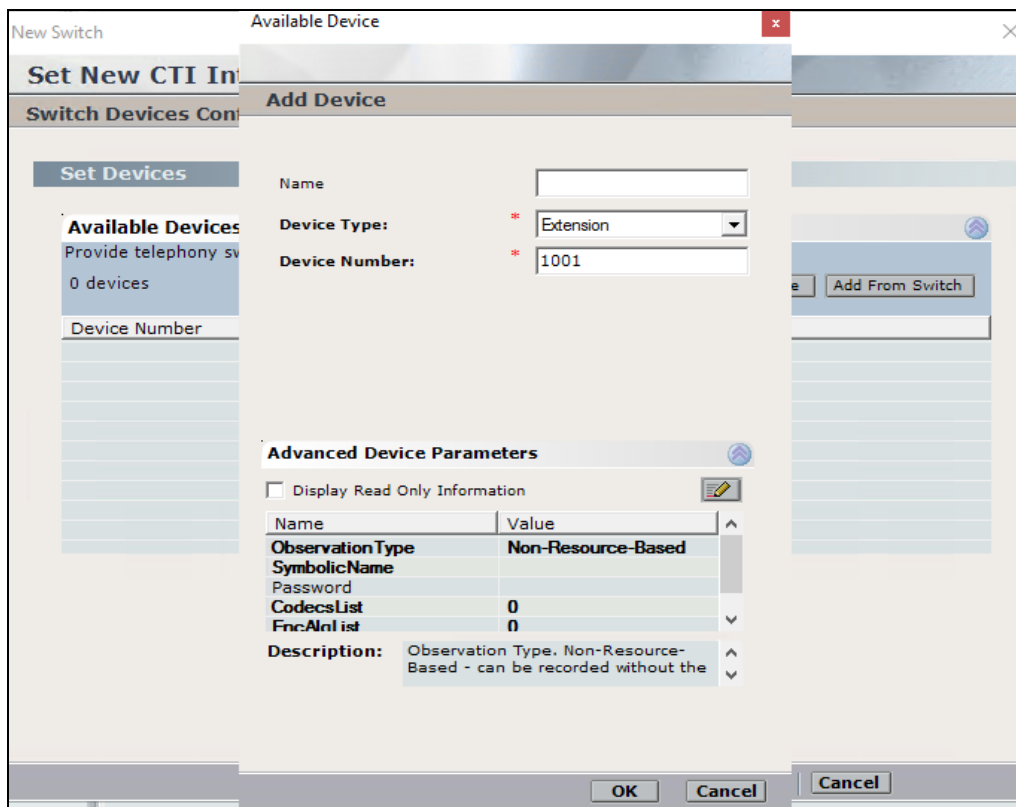
| IP/Hostname | CM Port |
|----------------|---------|
| NICEActive2012 | 62094 |
| | |
| | |

Back Next Cancel

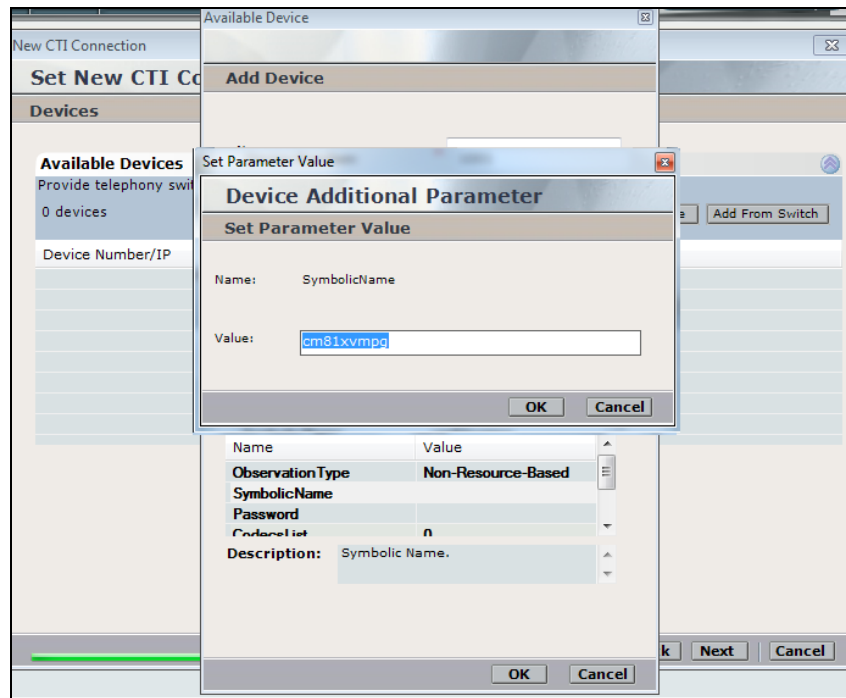
On the following screen, click on **Add**, to add the Communication Manager devices.



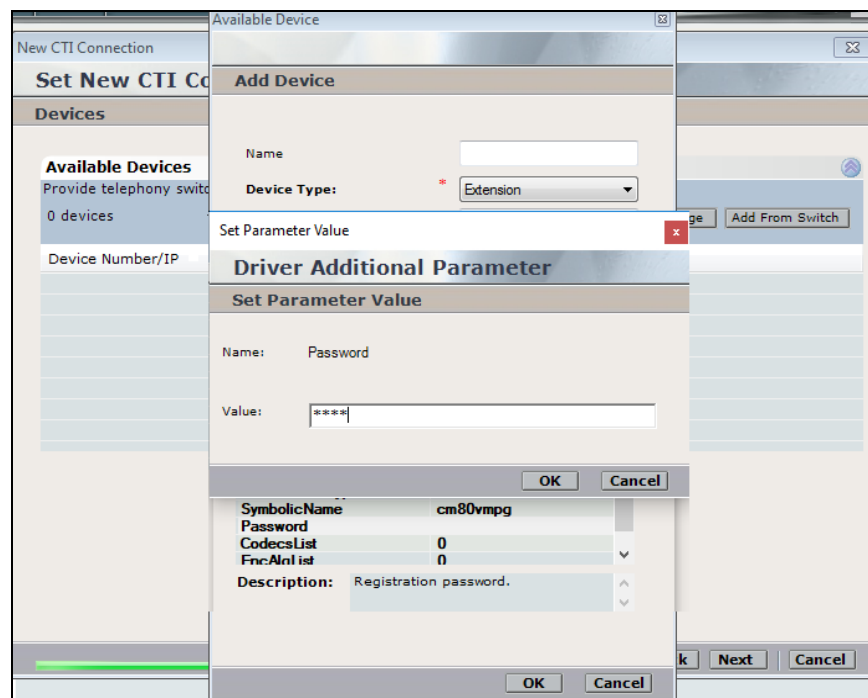
The **Device Type** should be **Extension** and insert the extension number of a phoneset that is to be recorded the example below showing extension **1001**. Expand **Advanced Device Parameters** and ensure that the **Value** for **Observation Type** is set to **Non-Resourced-Based**. Click on **OK** to continue.



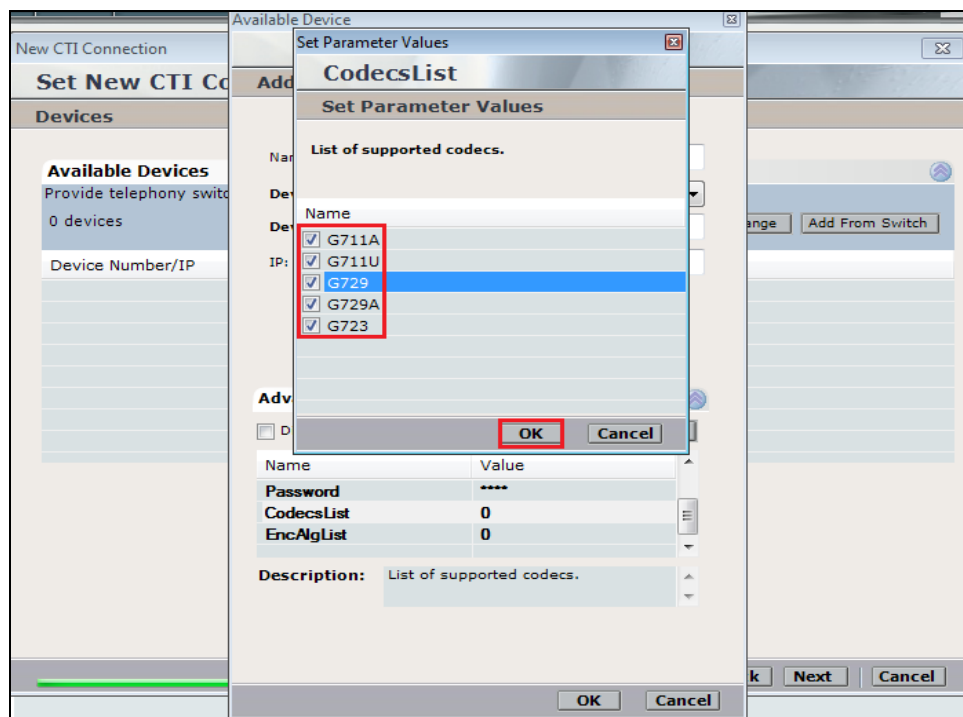
Enter the correct **Value** for **SymbolicName**. Double-click on **SymbolicName** to set the value. This should be the same as the switch name entered in **Section 6.2**.



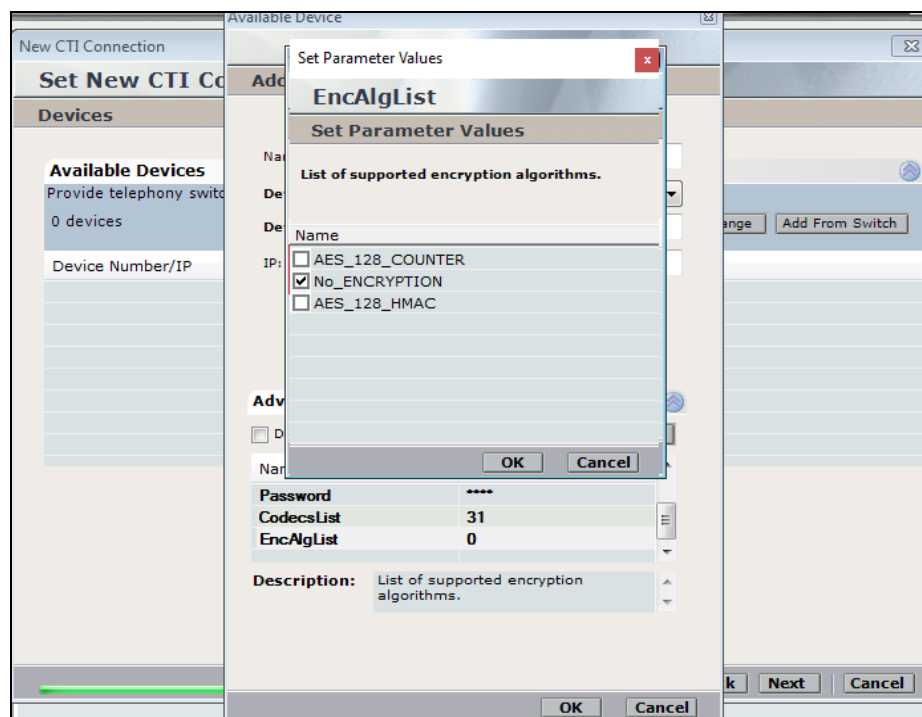
Enter the correct **Password** and note this is the password for the extension that is being added here. This is the station password which was entered during the creation of the station. A printout of an extension can be found in **Section 5.5** of these Application Notes.



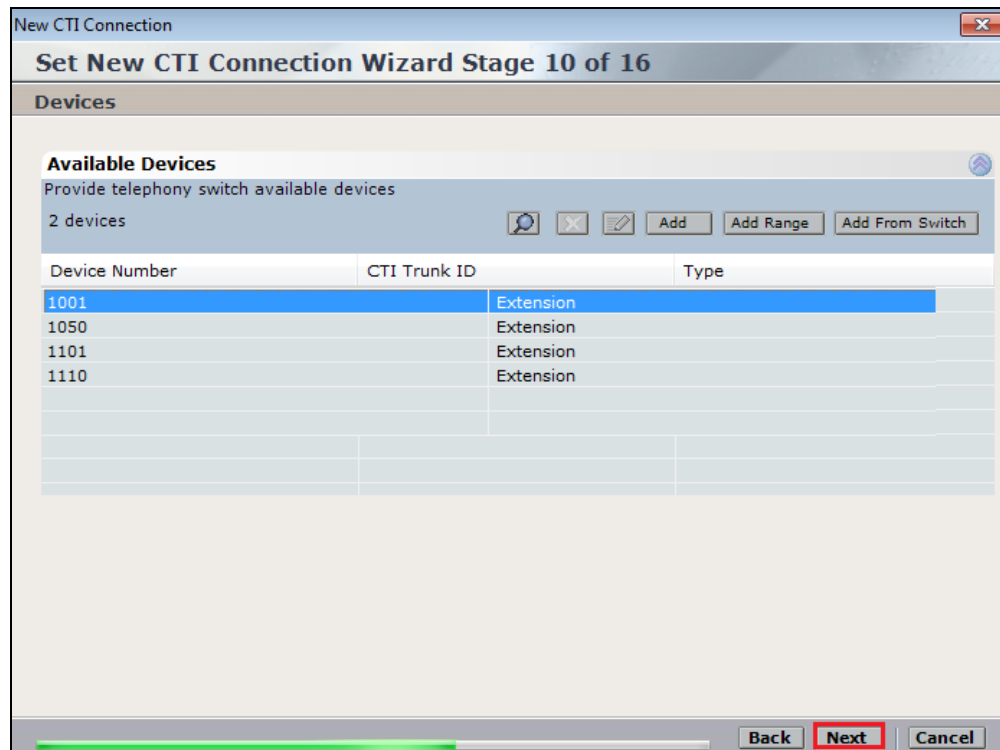
Double-click on **CodecsList** and ensure that all the values are ticked as shown below. Click on **OK** to continue.



Double-click on **EncAlgList** and since no SRTP was being recorded on this occasion **No_ENCRYPTION** was ticked. Click on **OK** to continue.



Click on **Next** to continue.



The screenshot shows the 'Set New CTI Connection Wizard Stage 10 of 16' window. The 'Devices' section is active, showing a list of available devices. The 'Available Devices' section includes a search icon, a close icon, and buttons for 'Add', 'Add Range', and 'Add From Switch'. Below this is a table with columns 'Device Number', 'CTI Trunk ID', and 'Type'. The table contains four rows of data, with the first row highlighted in blue. At the bottom of the window, there are 'Back', 'Next', and 'Cancel' buttons. The 'Next' button is highlighted with a red border.

| Device Number | CTI Trunk ID | Type |
|---------------|--------------|-----------|
| 1001 | | Extension |
| 1050 | | Extension |
| 1101 | | Extension |
| 1110 | | Extension |

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



The screenshot shows the 'Set New CTI Connection Wizard Stage 11 of 16' window. The 'Monitor' section is active, showing a list of available devices and a list of monitored devices. The 'Available Devices' section includes a search icon, a close icon, and buttons for '>>', '>', '<', and '<<'. Below this is a table with columns 'Device' and 'Type'. The table contains one row of data, with the first row highlighted in blue. At the bottom of the window, there are 'Back', 'Next', and 'Cancel' buttons. The 'Next' button is highlighted with a red border.

| Device | Type |
|--------|-----------|
| 1001 | Extension |

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

The screenshot shows a window titled "New CTI Connection" with a subtitle "Set New CTI Connection Wizard Stage 12 of 16". The main section is titled "Optional" and contains the text: "Select optional features relevant to integration. Some options may require further configuration." Below this text are four checkboxes: "SIP Trunk Correlation", "Rejected Devices", "Filter Calls", and "Call Flow Analysis". The "Call Flow Analysis" checkbox is checked and highlighted with a red rectangle. At the bottom right, there are three buttons: "Back", "Next" (highlighted with a red rectangle), and "Cancel". A green progress bar is visible at the bottom left.

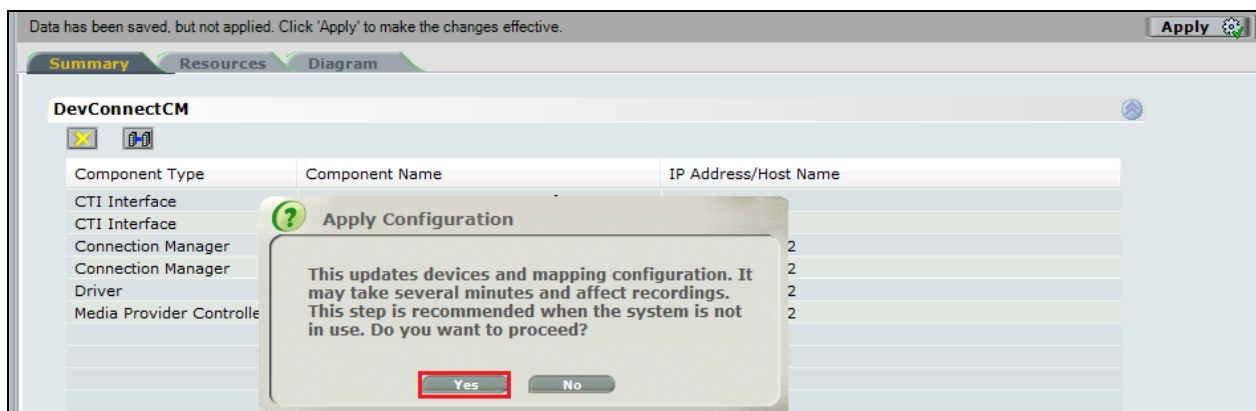
Select a different **Port** number as shown below **62095** is chosen simply because **62094** was already in use.

The screenshot shows a window titled "New CTI Connection" with a subtitle "Set New CTI Connection Wizard Stage 15 of 16". The main section is titled "Requirements" and contains the text: "The Interactions Center server selected already has a Connection Manager. Create a new Connection Manager, or select an existing one." Below this text are two radio buttons: "Create a new Connection Manager" (selected and highlighted with a red rectangle) and "Select available Connection Manager". Under "Create a new Connection Manager", there is a "Port:" label and a text box containing "62095", which is also highlighted with a red rectangle. Under "Select available Connection Manager", there is a "Ports in use:" label and a text box containing "62094". At the bottom right, there are three buttons: "Back", "Next" (highlighted with a red rectangle), and "Cancel". A green progress bar is visible at the bottom left.

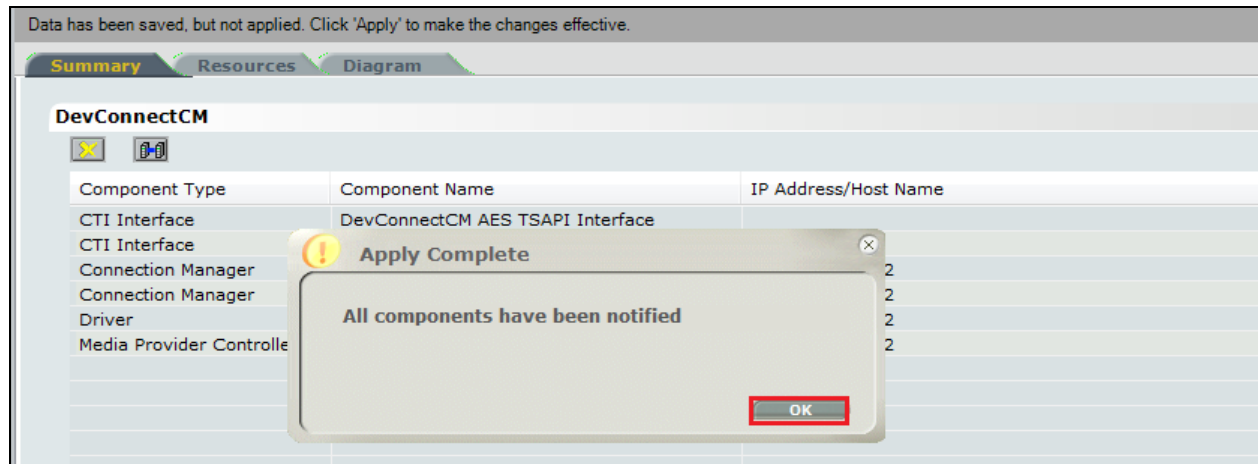
Click on **Finish** to complete the New CTI Wizard.



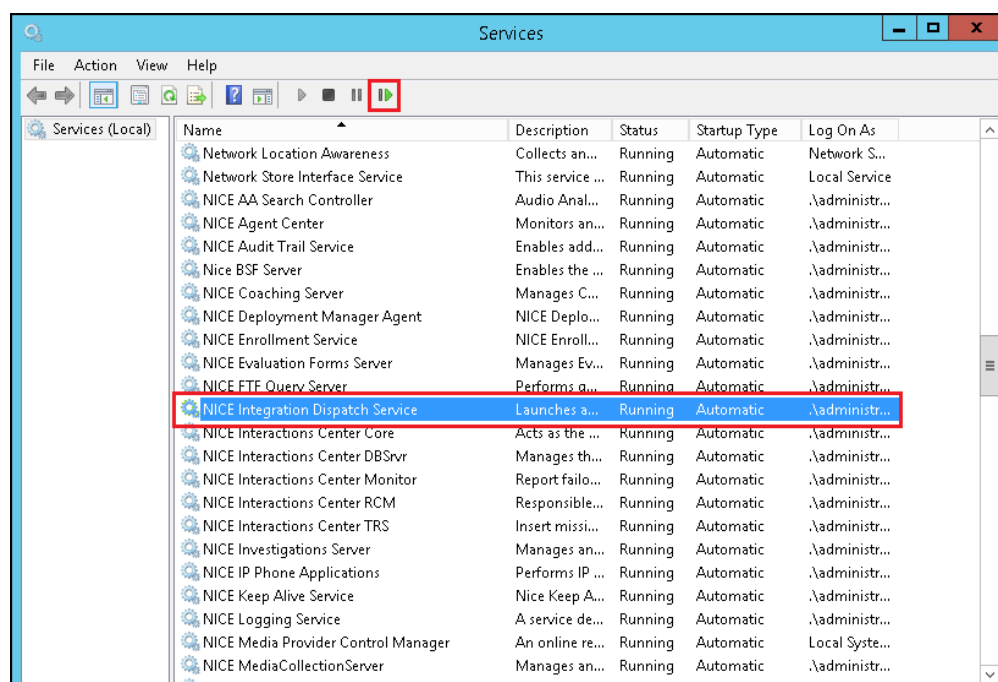
Click on **Apply** at the top right of the screen to save the new connection and click on **Yes** to proceed



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

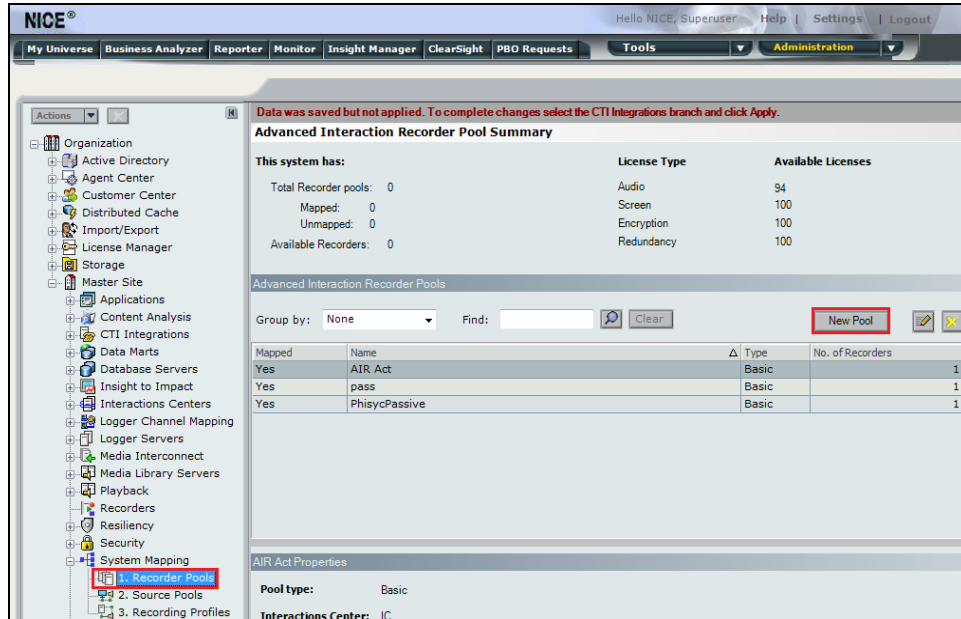


From the NICE Application Server, open **Services** and restart the **NICE Integration Dispatch Service**.

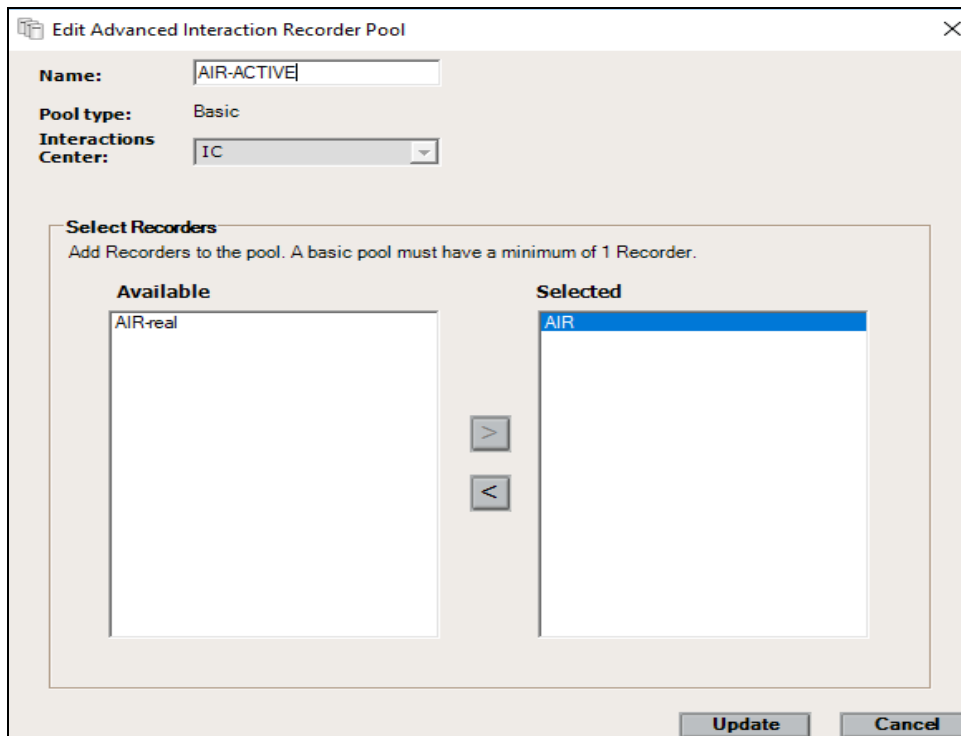


7.2. System Mapping

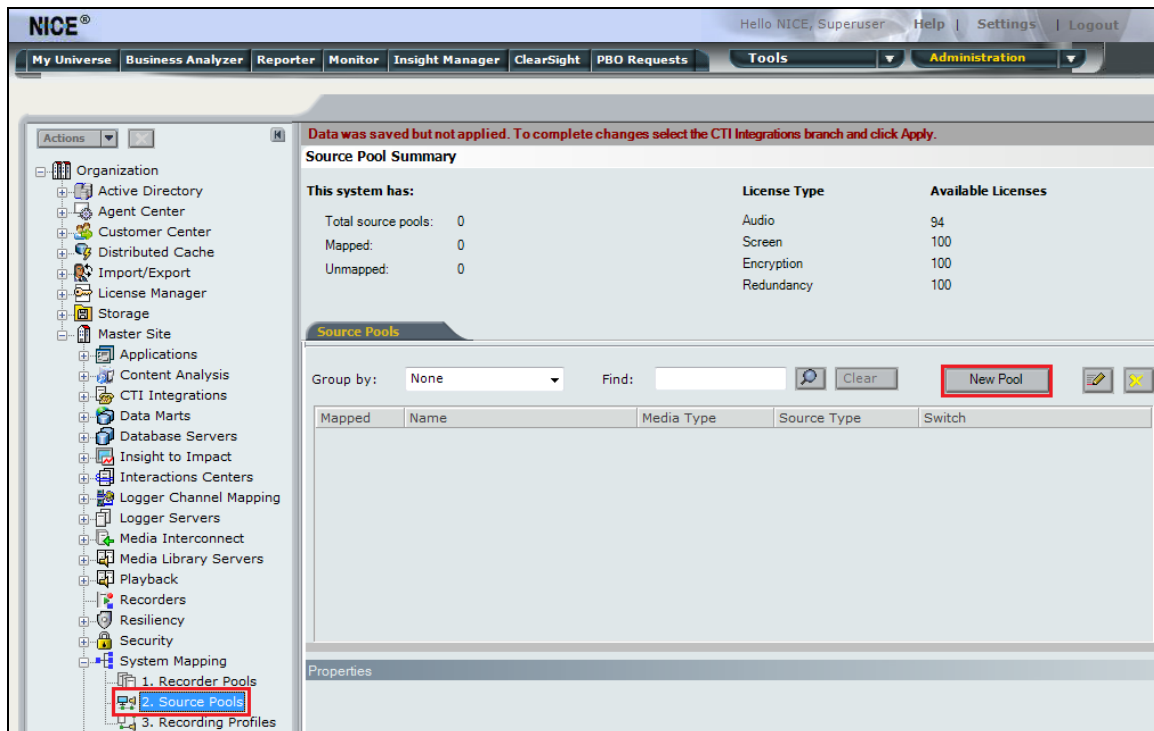
From the web browser navigate to **Master Site** → **System Mapping** → **Recorder Pools**. In the main window click on **New Pool**.



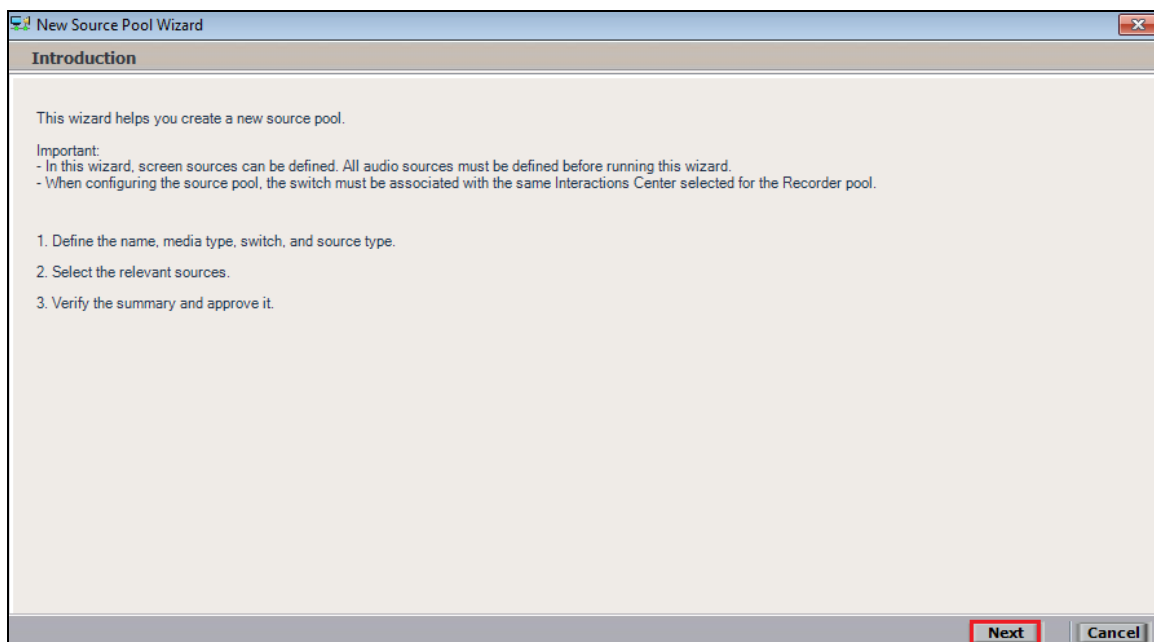
Enter a suitable **Name** for the **Recorder Pool** and select the **AIR** from the list of **Available Recorders** and click on **Update** to continue.



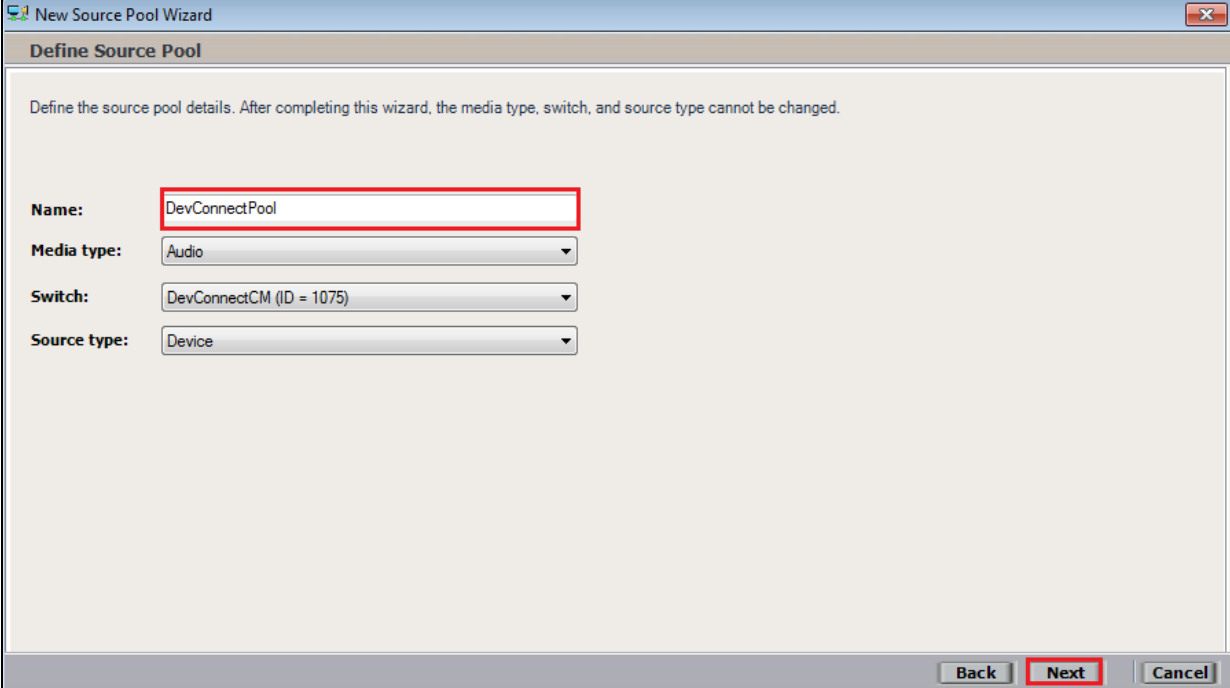
From the left navigation window select **Source Pools** and from the main window click on **New Pool**.



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

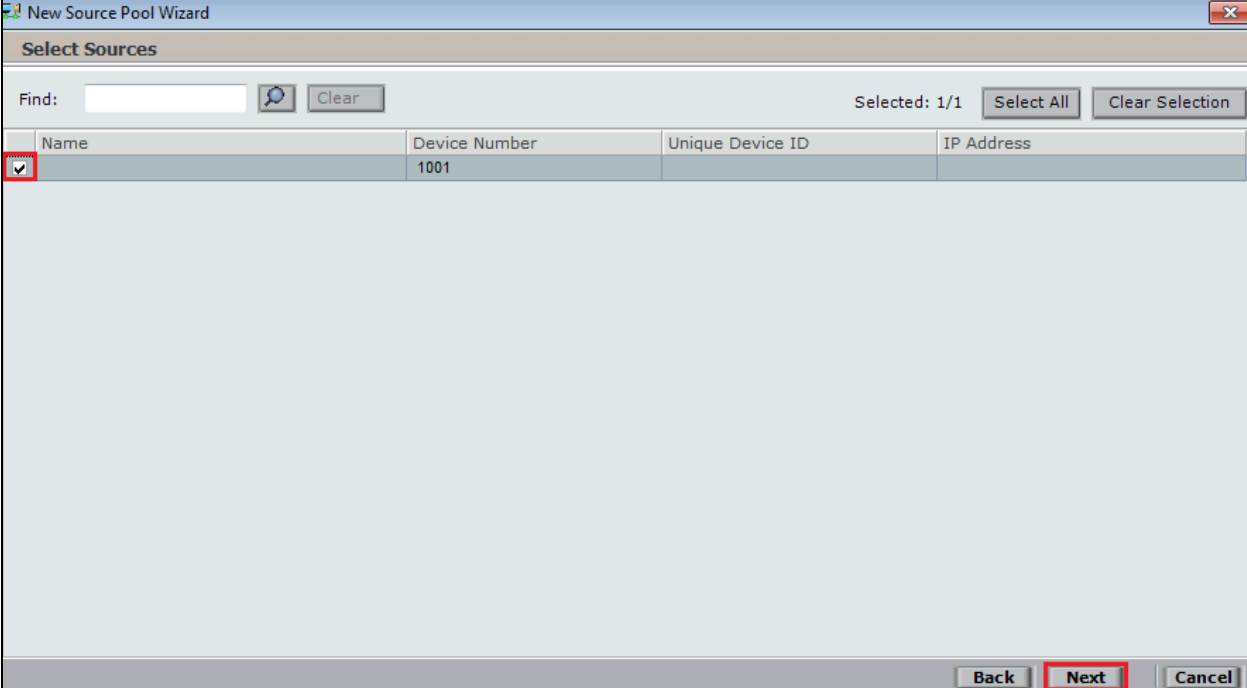


Enter a suitable **Name** and the other values were left as default. Click on **Next** to continue.



The screenshot shows the 'Define Source Pool' step of the 'New Source Pool Wizard'. The window title is 'New Source Pool Wizard'. The main heading is 'Define Source Pool'. Below the heading is a note: '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' (highlighted with a red box), and 'Cancel'.

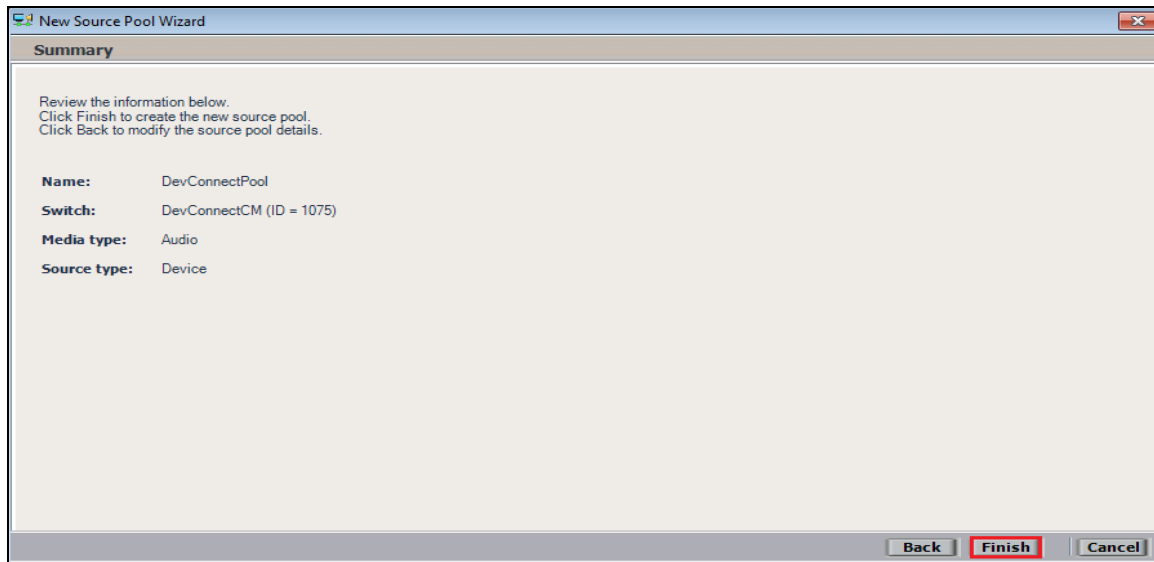
Select the extensions that were created in **Section 7.1**, note only one extension number is shown in the example below but this is not typical. Click on **Next** to continue.



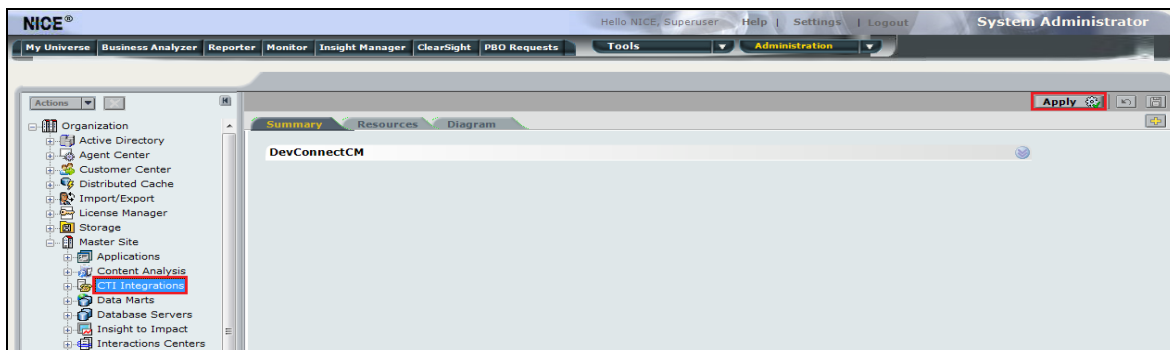
The screenshot shows the 'Select Sources' step of the 'New Source Pool Wizard'. The window title is 'New Source Pool Wizard'. The main heading is 'Select Sources'. There is a 'Find:' search bar with a magnifying glass icon and a 'Clear' button. To the right of the search bar, it says 'Selected: 1/1' and there are 'Select All' and 'Clear Selection' buttons. Below this is a table with four 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 '1001'. At the bottom right, there are three buttons: 'Back', 'Next' (highlighted with a red box), and 'Cancel'.

| | Name | Device Number | Unique Device ID | IP Address |
|-------------------------------------|------|---------------|------------------|------------|
| <input checked="" type="checkbox"/> | | 1001 | | |

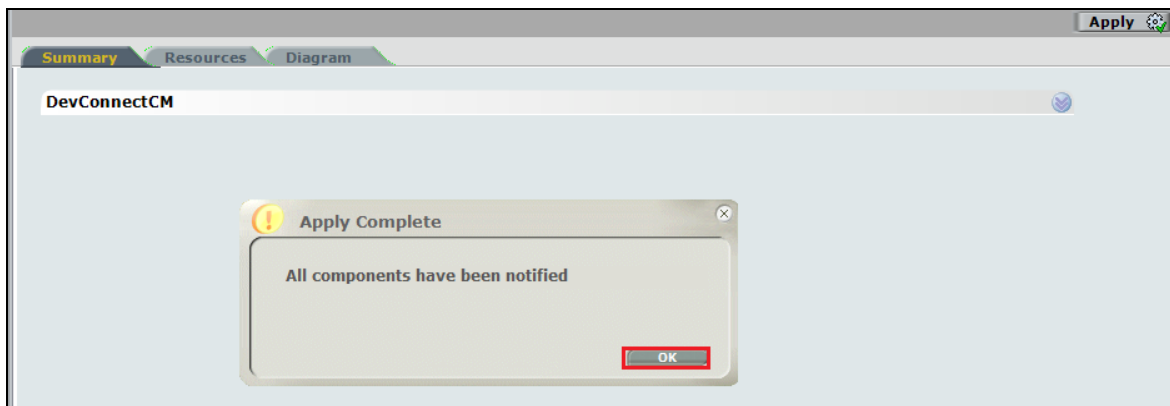
Click on **Finish** to complete the New Source Pool Wizard.



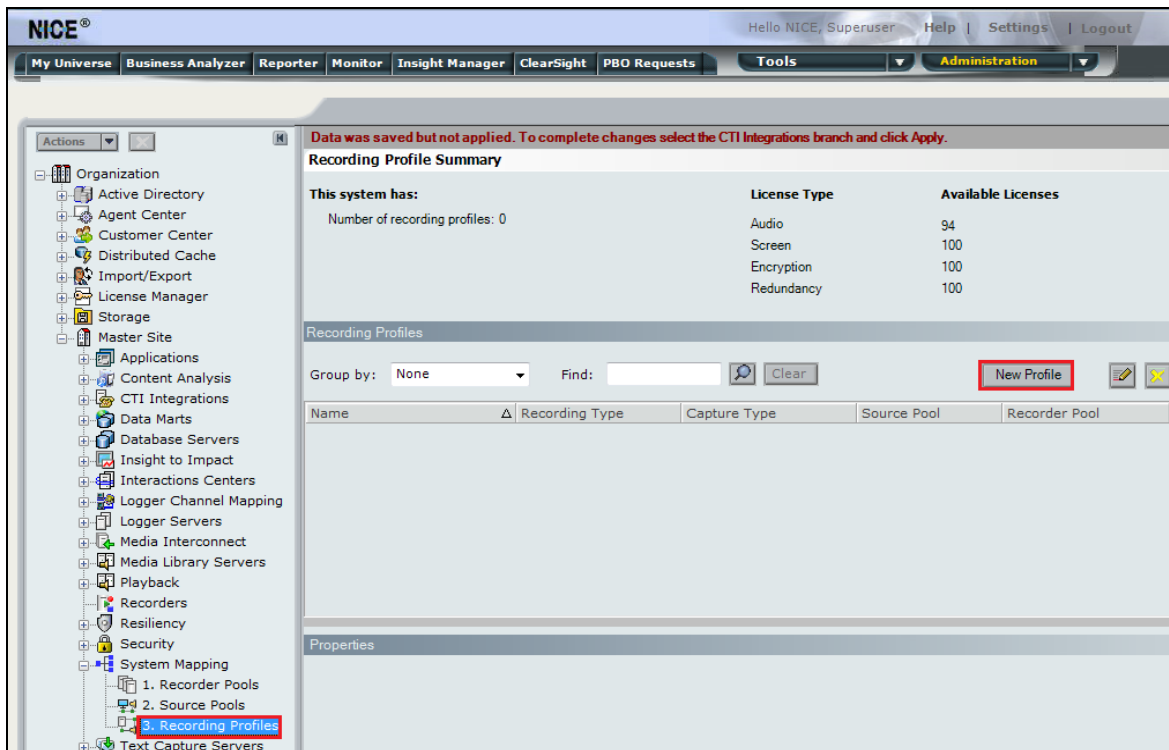
To implement these new changes, navigate to **Master Site → CTI Integrations** in the left window and in the main window click on **Apply** at the top right of the window.



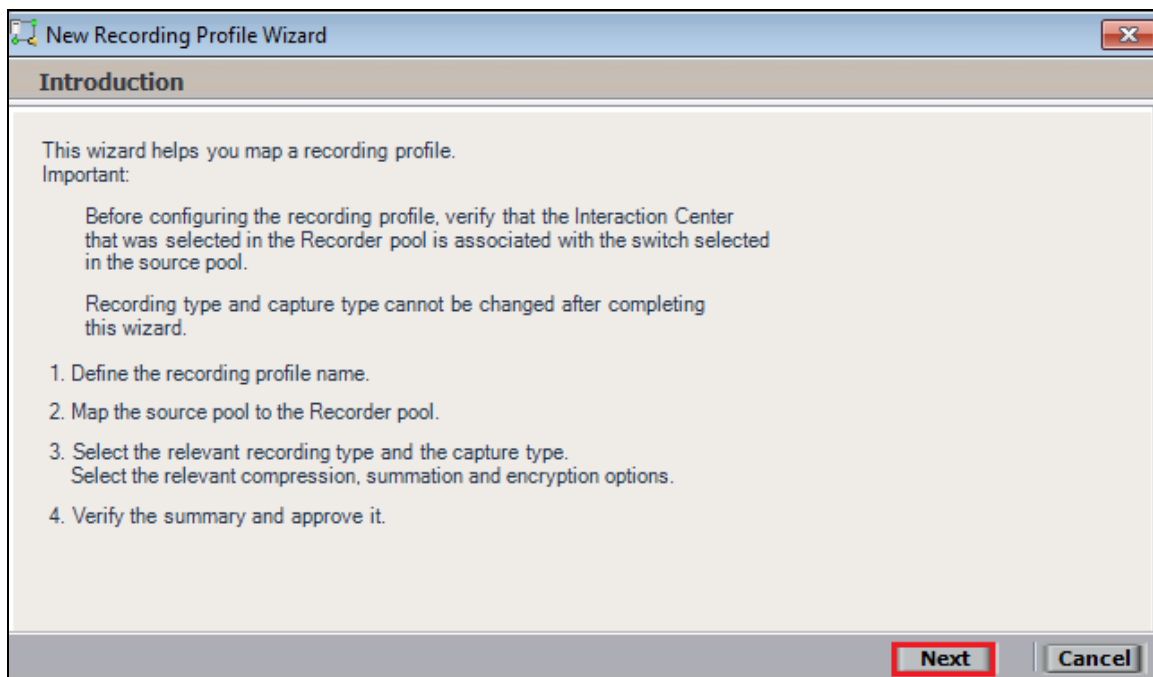
The following screen shows the changes were saved correctly. Click on **OK** to continue.



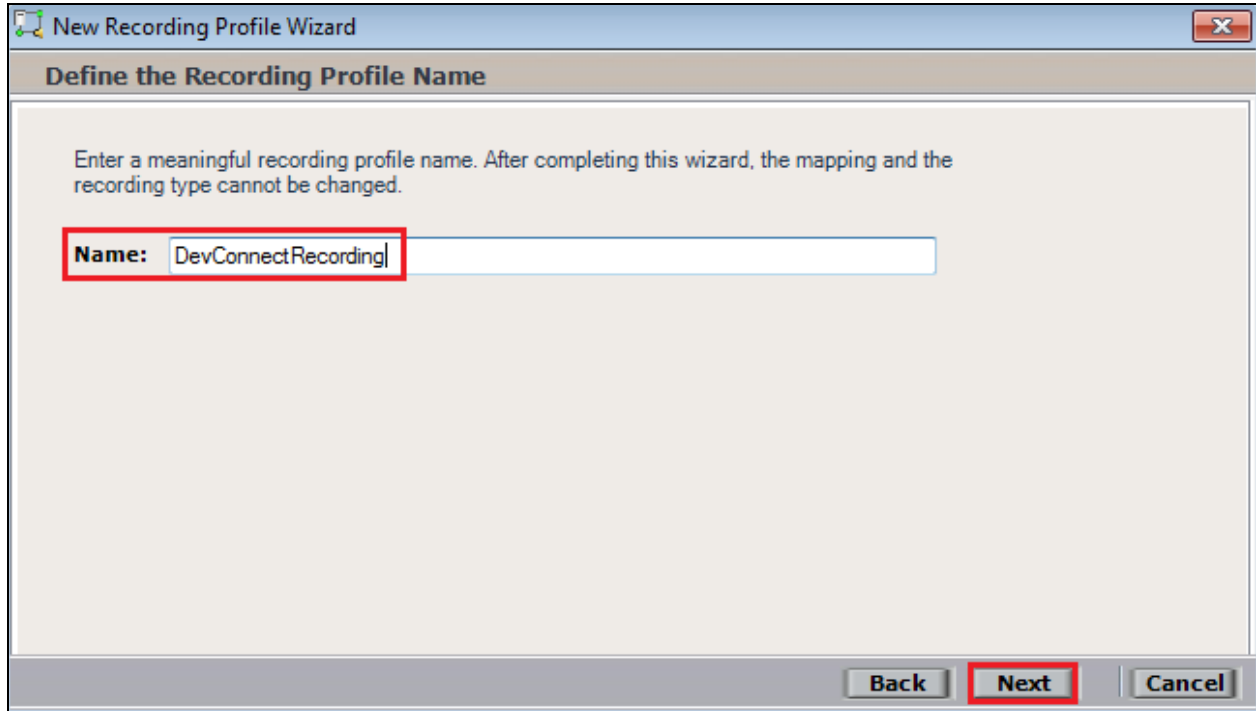
From the left window navigate to **Master Site** → **System Mapping** → **Recording Profiles** and in the main window click on **New Profile**.



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

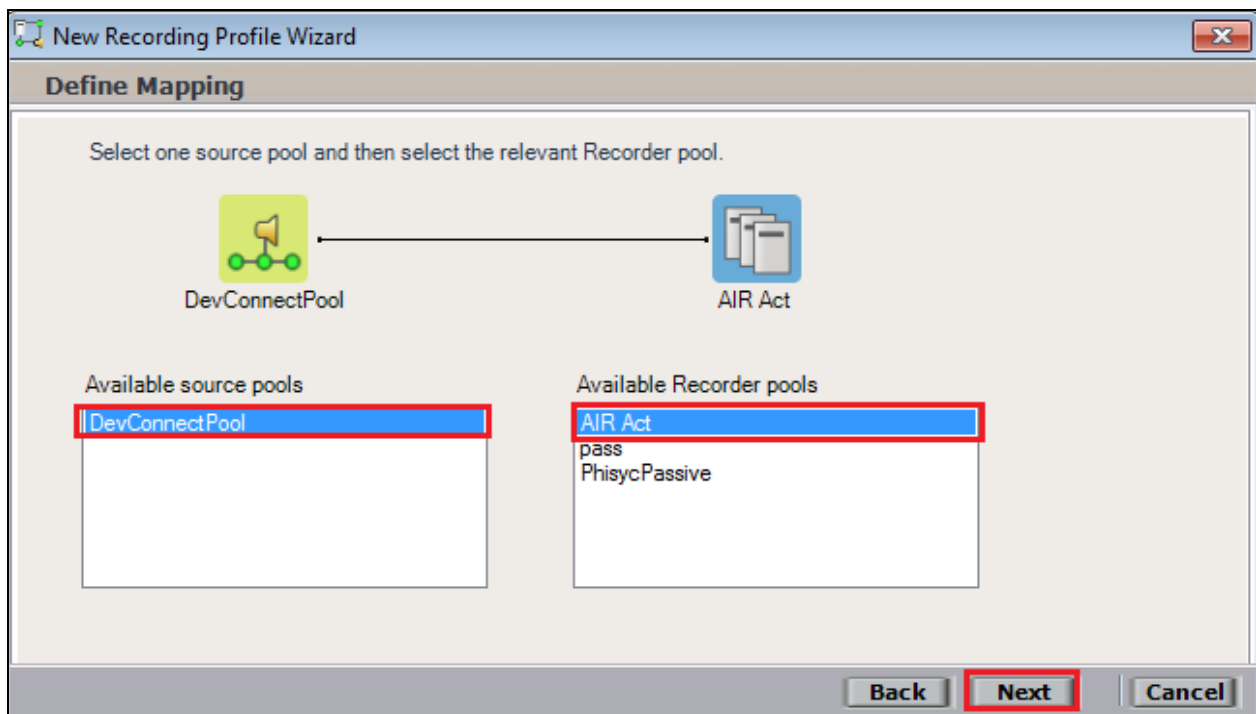


Enter a suitable **Name** for the Recording profile.



The screenshot shows the 'New Recording Profile Wizard' window, specifically the 'Define the Recording Profile Name' step. The window has a title bar with a close button. Below the title bar is a header area with the text 'Define the Recording Profile Name'. The main area contains a message: 'Enter a meaningful recording profile name. After completing this wizard, the mapping and the recording type cannot be changed.' Below this message is a text input field with the label 'Name:' and the text 'DevConnectRecording' entered. The input field is highlighted with a red border. At the bottom of the window are three buttons: 'Back', 'Next', and 'Cancel'. The 'Next' button is highlighted with a red border.

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



The screenshot shows the 'New Recording Profile Wizard' window, specifically the 'Define Mapping' step. The window has a title bar with a close button. Below the title bar is a header area with the text 'Define Mapping'. The main area contains a message: 'Select one source pool and then select the relevant Recorder pool.' Below this message is a diagram showing a source pool icon (a green square with a yellow circle and a green line) labeled 'DevConnectPool' and a recorder pool icon (a blue square with a white circle and a blue line) labeled 'AIR Act'. Below the diagram are two lists: 'Available source pools' and 'Available Recorder pools'. The 'Available source pools' list contains 'DevConnectPool' and is highlighted with a red border. The 'Available Recorder pools' list contains 'AIR Act', 'pass', and 'PhisycPassive', and is also highlighted with a red border. At the bottom of the window are three buttons: 'Back', 'Next', and 'Cancel'. The 'Next' button is highlighted with a red border.

For total recording i.e., the recording of all calls, select **Total** as the **Recording type**. For **Capture type** ensure that **Active DMCC MR Stereo** and **By Device** is selected beside it. **Audio Compression** is selected as default and can be left like this. Click on **Next** to continue.

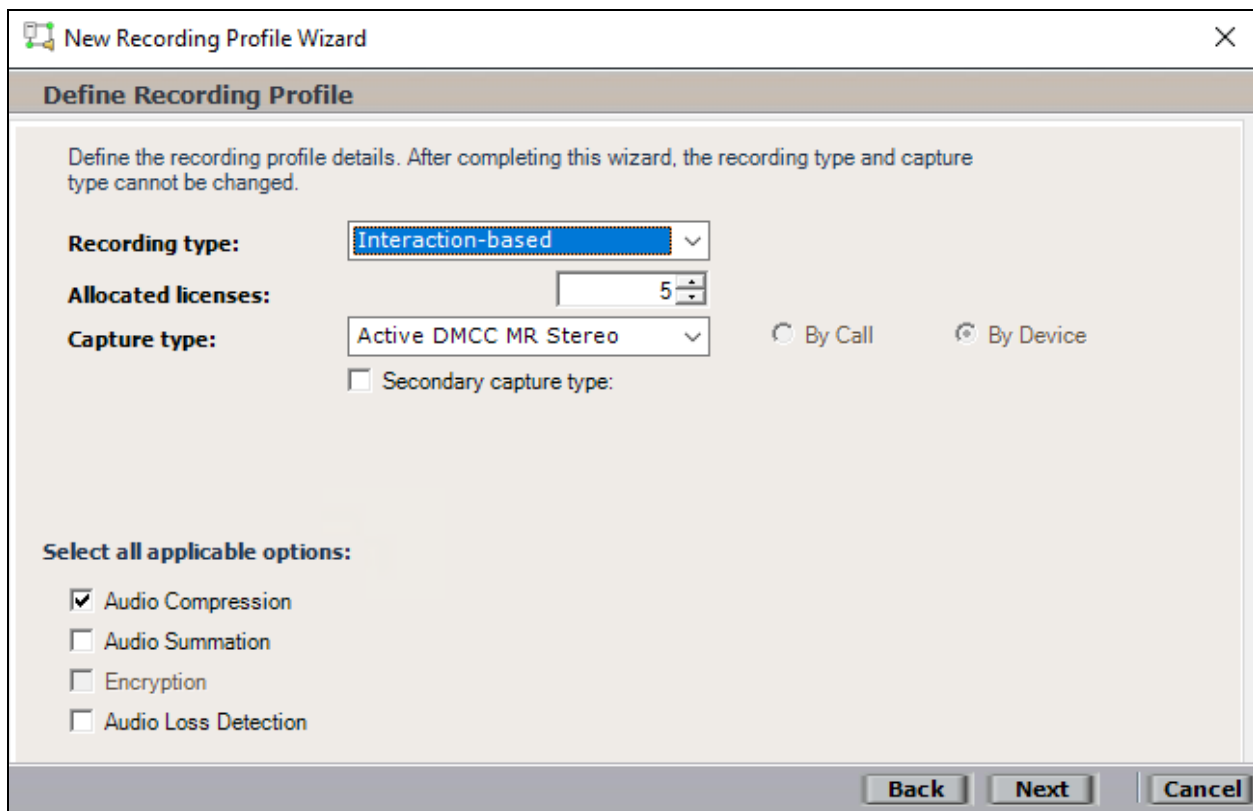
The screenshot shows a window titled "New Recording Profile Wizard" with a close button (X) in the top right corner. The main heading is "Define Recording Profile". Below this, a note states: "Define the recording profile details. After completing this wizard, the recording type and capture type cannot be changed." The form contains the following fields and options:

- Recording type:** A dropdown menu with "Total" selected.
- Allocated licenses:** A text field containing "Determined by the number of sources in the source pool".
- Capture type:** A dropdown menu with "Active DMCC MR Stereo" selected.
- Two radio buttons: "By Call" (unselected) and "By Device" (selected).
- A checkbox labeled "Secondary capture type:" which is unchecked.
- A section titled "Select all applicable options:" containing four checkboxes:
 - ☒ Audio Compression
 - ☐ Audio Summation
 - ☐ Encryption
 - ☐ Audio Loss Detection

At the bottom of the window are three buttons: "Back", "Next", and "Cancel".

Note: Avaya would recommend that **Total** “recording type” is used as it is not recommended to have recorders registering and unregistering to cope with an “interaction-based” type of recording.

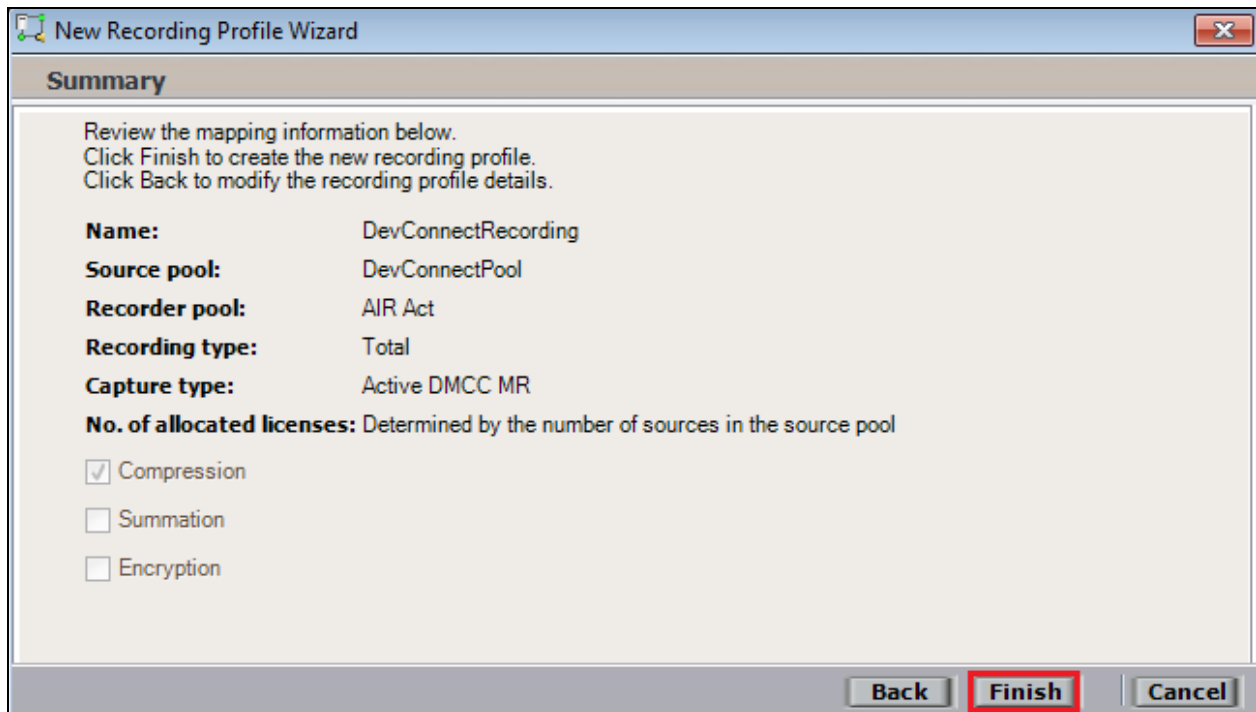
Interaction-based recording can be configured by selecting **Interaction-based** as the **Recording type** and **Active DMCC MR Stereo** as the **Capture type** and **By Device** is selected beside it. **Audio Compression** is selected as default and can be left like this. Click on **Next** to continue.



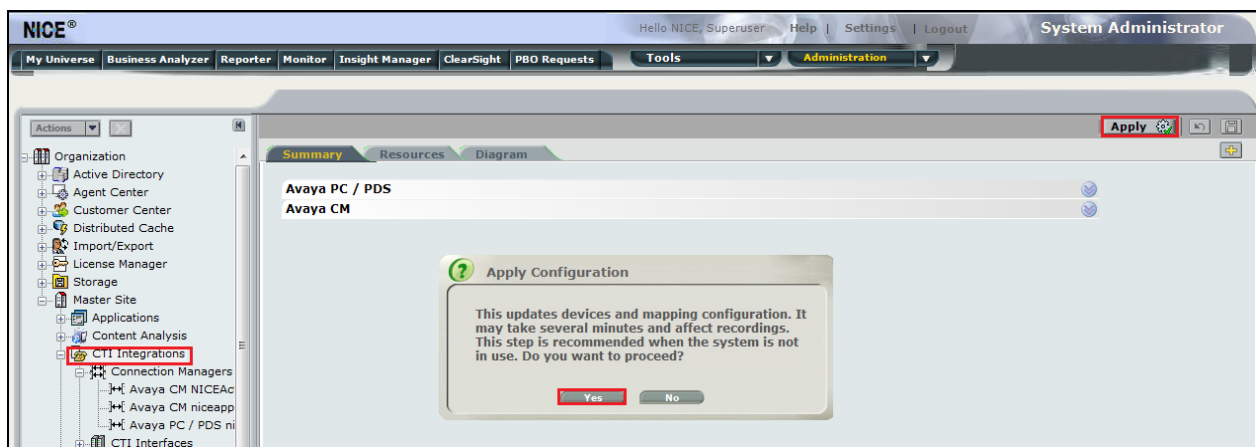
The image shows a 'New Recording Profile Wizard' dialog box with a title bar containing a small icon and a close button. The main area is titled 'Define Recording Profile' and contains the following elements:

- A text instruction: 'Define the recording profile details. After completing this wizard, the recording type and capture type cannot be changed.'
- 'Recording type:' dropdown menu set to 'Interaction-based'.
- 'Allocated licenses:' spinner box set to '5'.
- 'Capture type:' dropdown menu set to 'Active DMCC MR Stereo'.
- Radio buttons for 'By Call' (unselected) and 'By Device' (selected).
- An unchecked checkbox for 'Secondary capture type:'.
- A section titled 'Select all applicable options:' with four checkboxes:
 - ☒ Audio Compression
 - ☐ Audio Summation
 - ☐ Encryption
 - ☐ Audio Loss Detection
- At the bottom, three buttons: 'Back', 'Next', and 'Cancel'.

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



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 Multiple Registration recording.

8. Verification Steps

This section provides the steps that can be taken to verify correct configuration of the NICE Engage Platform and 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 checked, check the connection between Communication Manager and AES 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 | 11 | no | aes81vmpg | established | 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**.

| Status Status and Control TSAPI Service Summary | | | | | | | | | | | |
|--|------|-------------|--------------------|-------------|--------------------------|--------|----------------|--------------|----------------|------------------|-------------|
| Home Help Logout | | | | | | | | | | | |
| <div>▶ AE Services</div> <div>▶ Communication Manager Interface</div> <div>▶ High Availability</div> <div>▶ Licensing</div> <div>▶ Maintenance</div> <div>▶ Networking</div> <div>▶ Security</div> <div>▼ Status</div> <div>Alarm Viewer</div> <div>▶ Logs</div> <div>▶ Log Manager</div> <div>▼ Status and Control</div> <div>■ CVLAN Service Summary</div> | | | | | | | | | | | |
| TSAPI Link Details | | | | | | | | | | | |
| <input type="checkbox"/> Enable page refresh every 60 seconds | | | | | | | | | | | |
| | Link | Switch Name | Switch CTI Link ID | Status | Since | State | Switch Version | Associations | Msgs to Switch | Msgs from Switch | Msgs Period |
| <input checked="" type="radio"/> | 1 | cm81xvmpg | 1 | Talking | Sat May 22 18:25:51 2021 | Online | 18 | 8 | 21 | 22 | 30 |
| <input type="radio"/> | 2 | cm81large | 1 | Switch Down | Wed Apr 14 15:25:43 2021 | Online | 18 | 0 | 0 | 0 | 30 |
| <div>Online Offline</div> | | | | | | | | | | | |
| For service-wide information, choose one of the following: | | | | | | | | | | | |
| <div>TSAPI Service Status TLink Status User Status</div> | | | | | | | | | | | |

Clicking on **User Status** from the screen on the previous page should display something similar to that shown below, where the NICE user and corresponding **Tlink Name** are shown.

- AE Services
- Communication Manager Interface
- High Availability
- Licensing
- Maintenance
- Networking
- Security
- Status**
 - Alarm Viewer
 - Logs
 - Log Manager
 - Status and Control**
 - CVLAN Service Summary
 - DLG Services Summary
 - DMCC Service Summary
 - Switch Conn Summary
 - TSAPI Service Summary

CTI User Status

☐ Enable page refresh every 60 seconds

CTI Users: All Users Submit

Open Streams: 5
Closed Streams: 25

Open Streams

| Name | Time Opened | Time Closed | Tlink Name |
|------------------------|---------------------------------|-------------|---------------------------------|
| nice1 | Thu 03 Jun 2021 04:00:07 PM IST | | AVAYA#CM81XVMPG#CSTA#AES81XVMPG |
| DMCCLCSUserDoNotModify | Wed 14 Apr 2021 03:27:12 PM IST | | AVAYA#CM81XVMPG#CSTA#AES81XVMPG |
| DMCCLCSUserDoNotModify | Wed 14 Apr 2021 03:27:12 PM IST | | AVAYA#CM81LARGE#CSTA#AES81XVMPG |
| DMCCLCSUserDoNotModify | Wed 14 Apr 2021 03:27:13 PM IST | | AVAYA#CM81XVMPG#CSTA#AES81XVMPG |
| DMCCLCSUserDoNotModify | Wed 14 Apr 2021 03:27:13 PM IST | | AVAYA#CM81LARGE#CSTA#AES81XVMPG |

Show Closed Streams Close All Opened Streams Back

8.3. Verify DMCC link on AES

Verify the status of the DMCC link by selecting **Status** → **Status and Control** → **DMCC Service Summary** to display the **DMCC Service Summary – Session Summary** screen. The screen below shows that the user **NICE** is connected from the IP address **10.10.40.121**, which is the NICE Application server.

Status | Status and Control | DMCC Service Summary

Home | Help | Logout

- AE Services
- Communication Manager Interface
- High Availability
- Licensing
- Maintenance
- Networking
- Security
- Status**
 - Alarm Viewer
 - Logs
 - Log Manager
 - Status and Control**
 - CVLAN Service Summary
 - DLG Services Summary
 - DMCC Service Summary**
 - Switch Conn Summary
 - TSAPI Service Summary

DMCC Service Summary - Session Summary

Please do not use back button

☐ Enable page refresh every 60 seconds

Session Summary [Device Summary](#)
Generated on Fri Dec 07 13:35:57 GMT 2018

Service Uptime: 15 days, 4 hours 19 minutes
Number of Active Sessions: 1
Number of Sessions Created Since Service Boot: 20
Number of Existing Devices: 10
Number of Devices Created Since Service Boot: 94

| | Session ID | User | Application | Far-end Identifier | Connection Type | # of Associated Devices |
|--------------------------|--------------------------------------|------|-------------|--------------------|-----------------|-------------------------|
| <input type="checkbox"/> | A5DD1D50223FFC1F4 B7C472898A98AF2-19 | NICE | | 10.10.40.121 | XML Unencrypted | 10 |

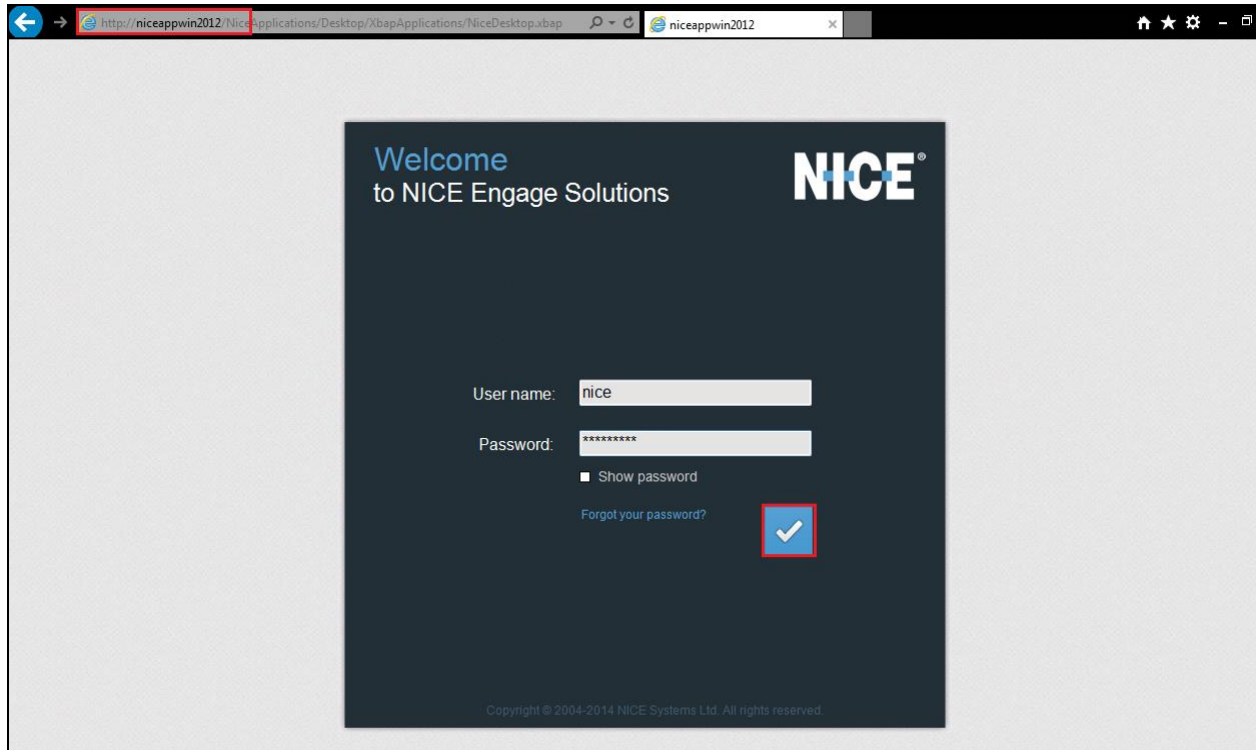
Terminate Sessions Show Terminated Sessions

Item 1-1 of 1
1 Go

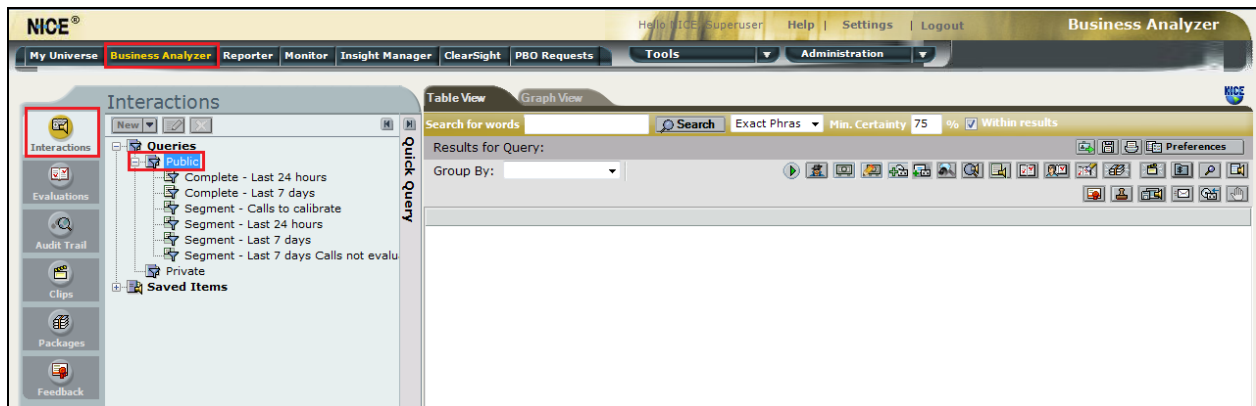
8.4. Verify calls are being recorded

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.

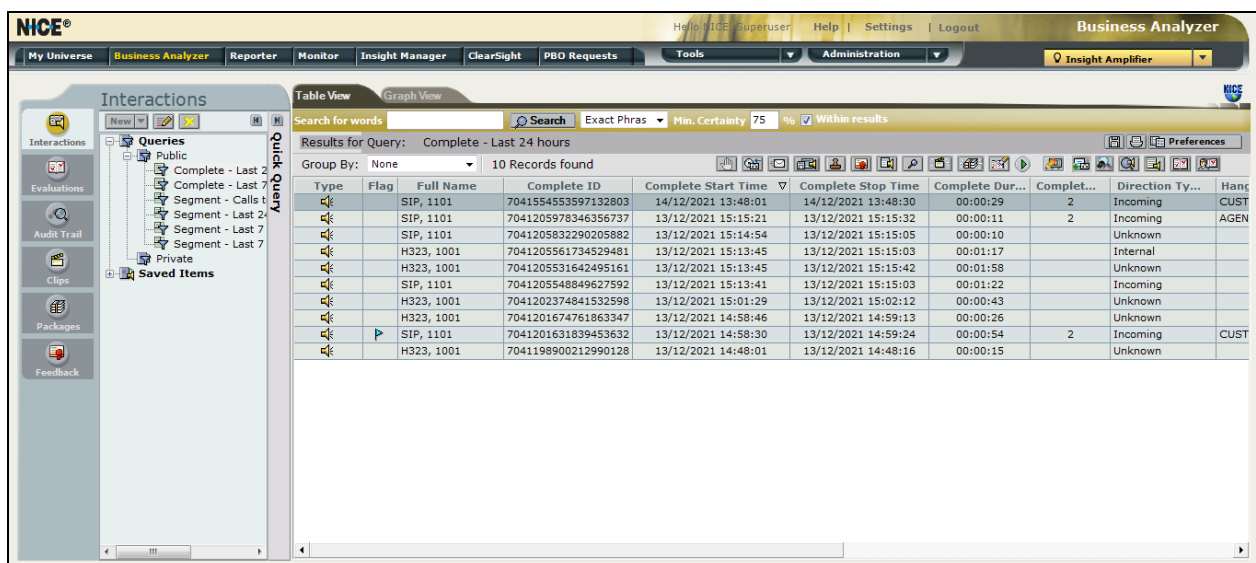
Open a browser session to the NICE Application Server as is shown below. Enter the proper credentials and click on **Login**.



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

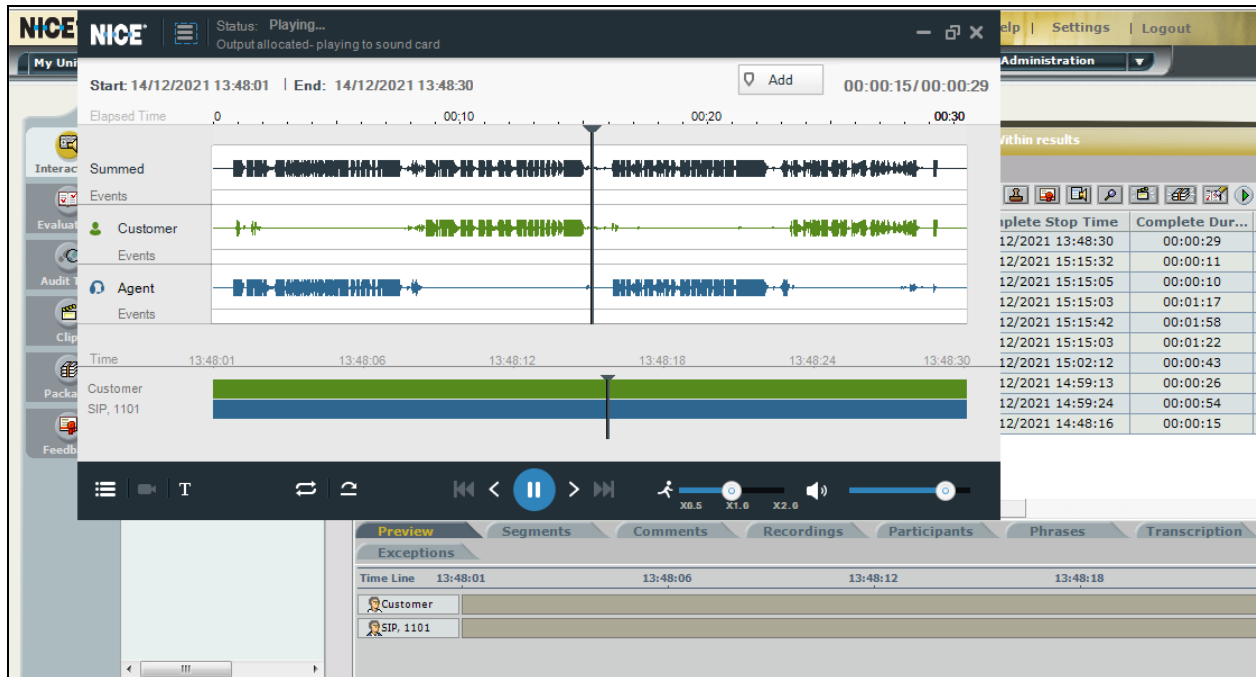


Click on **Complete – Last 24 hours**. This should reveal all the recordings that took place over the previous 24 hours. Select the required recording from the list and double-click on this to play the recording.



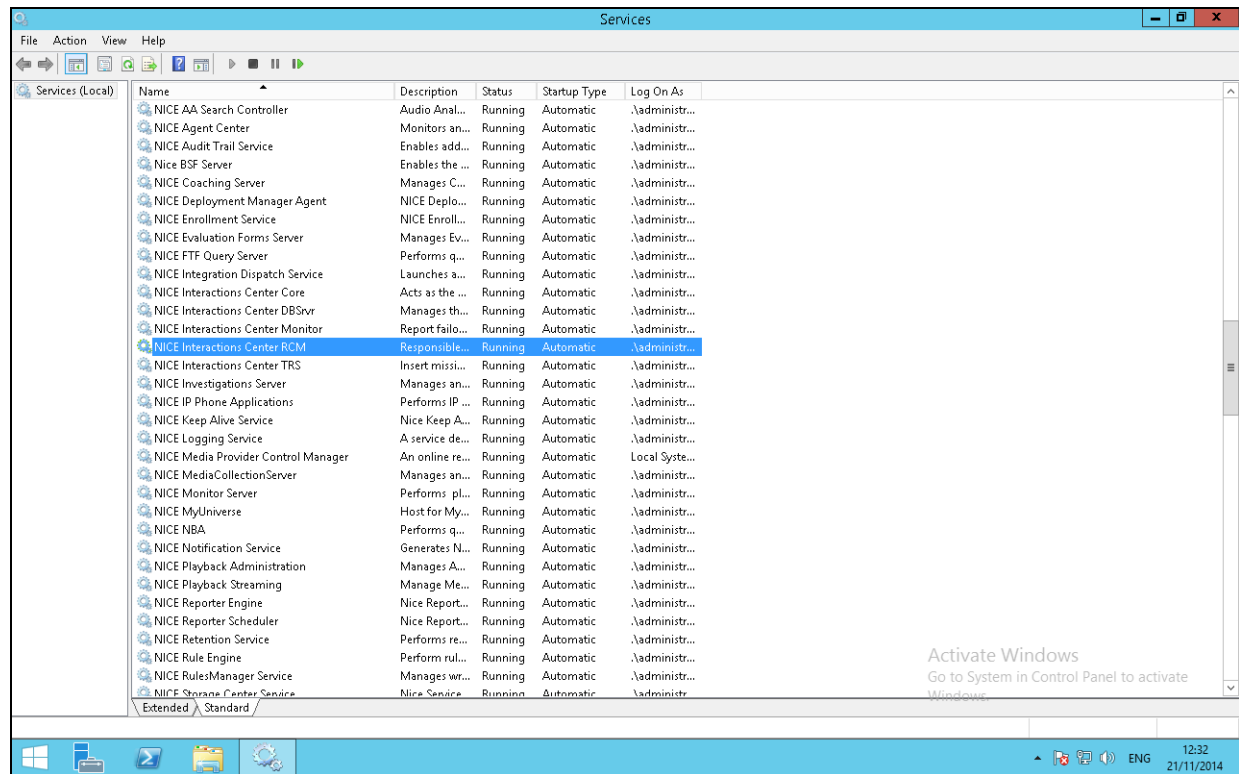
The NICE player is opened, and the recording is presented for playback. Click on the **Play/Pause** icon highlighted below to play back the recording.

Note: The recording below shows two separate streams in stereo, with the **Customer** on one side and the **Agent** on the other.



8.5. 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 Recorder 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 R7.1 to successfully interoperate with Avaya Aura® Communication Manager R8.1 using Avaya Aura® Application Enablement Services R8.1 to connect to using DMCC Multiple Registration to record calls in stereo. All feature functionality and serviceability test cases were completed successfully with 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 8.1
- [4] *Avaya Aura® Session Manager Overview and Specifications*

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

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