

Avaya Solution & Interoperability Test Lab

Application Notes for Enghouse Interactive Communications Center 10.0 with Avaya Aura® Communication Manager 7.1 using Avaya Aura® Application Enablement Services 7.1 – Issue 1.0

Abstract

These Application Notes describe the configuration steps required for Enghouse Interactive Communications Center 10.0 to interoperate with Avaya Aura® Communication Manager 7.1 using Avaya Aura® Application Enablement Services 7.1. Enghouse Interactive Communications Center is a multi-channel and multi-contact solution that can handle voice, fax, web, and email contacts.

The compliance testing focused on the voice integration with Avaya Aura® Communication Manager via the Avaya Aura® Application Enablement Services Telephony Services Application Programming Interface and Device, Media, and Call Control interface.

Readers should pay attention to **Section 2**, in particular the scope of testing as outlined in **Section 2.1** as well as any 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 required for Enghouse Interactive Communications Center (EICC) 10.0 to interoperate with Avaya Aura® Communication Manager 7.1 using Avaya Aura® Application Enablement Services 7.1. EICC is a multi-channel and multi-contact solution that can handle voice, fax, web, and email contacts.

The compliance testing focused on the voice integration with Communication Manager via the Application Enablement Services Telephony Services Application Programming Interface (TSAPI) and Device, Media, and Call Control (DMCC) interface.

In the compliance testing, agents and supervisors were configured as station users on Communication Manager, and have desktop computers running the Enghouse Interactive TouchPoint client application. The ACD functionality such as log in/out, work modes, queuing, and announcements were provided by EICC.

The TSAPI interface was used by EICC to monitor agent and supervisor station extensions, provide screen pops and call control from agent desktops, route incoming calls using adjunct routing capability, and support enable/disable of call forwarding and message waiting lamp using set value capability. In addition, TSAPI single step conference was used to support the supervisor monitor feature, which can be activated from the supervisor desktop running the TouchPoint application.

The DMCC interface was used by EICC to support voicemail, announcement, and basic call recording features via virtual IP softphones. The virtual IP softphones were registered by EICC with Communication Manager. Voicemail and announcement calls were redirected to available virtual IP softphones to terminate to EICC, and recording was accomplished by intruding a virtual IP softphone via TSAPI single step conference onto the active call to pick up media for recording.

2. General Test Approach and Test Results

The feature test cases were performed both automatically and manually. Upon start of the EICC application, the application automatically used TSAPI to query device name, requested device monitoring, and registered for VDN routing. The application also automatically used DMCC to register the virtual IP softphones.

For the manual part of the testing, incoming calls were made to the general routing VDNs. The EICC server used query results and event reports to track agent states, and specified calls to be routed to available agents or to call treatment VDNs. Manual call controls from the TouchPoint client application were exercised to verify call control features such as answering and transferring of calls.

Voicemail was tested by not answering call at the agent, and have the call covered to EICC with proper leaving of voice message and activation of agent message waiting lamp. Manual call was then made from the agent to the voicemail VDN to retrieve voice message and verify proper deactivation of message waiting lamp.

The serviceability test cases were performed manually by disconnecting and reconnecting the Ethernet connection to the EICC server and clients.

The verification of tests included human checking of proper states at the telephones, and of capturing and analyzing the TSAPI and DMCC message traces from the EICC server.

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 Application Enablement Services and EICC did not include use of any specific encryption features as requested by Enghouse Interactive.

2.1. Interoperability Compliance Testing

The interoperability compliance test included feature and serviceability testing.

The feature testing focused on verifying the following on EICC:

- Use of TSAPI query service to query device names.
- Use of TSAPI event report service to monitor agents, supervisor, and virtual IP softphones.
- Use of TSAPI routing service to route incoming calls.
- Use of TSAPI set value service to activate/deactivate call forwarding and message waiting lamp.
- Use of TSAPI call control service to support manual call control actions initiated from TouchPoint, call control for virtual IP softphones, and adding virtual IP softphones to existing calls for media capture.
- Use of DMCC registration service to register and un-register the virtual IP softphones.
- Proper handling of call scenarios involving screen pop, inbound, outbound, ACD, non-ACD, drop, hold/reconnect, voicemail, message waiting lamp, blind/attended transfer, attended conference, call forwarding, supervisor monitor, multiple agents, multiple calls, queuing, send DTMF, long duration, and recording of basic calls.

The serviceability testing focused on verifying the ability of EICC to recover from adverse conditions, such as disconnecting/reconnecting Ethernet connection to EICC server and clients.

2.2. Test Results

All test cases were executed. The following were observations on EICC from the compliance testing.

- EICC created one DMCC version per virtual IP softphone by design.
- For the attended conference scenario, after the PSTN drops, one of the remaining agent's Phone Calls section reflected his/her name instead of name of the other agent.

2.3. Support

Technical support on EICC can be obtained through the following:

• **Phone:** (800) 513-2810

Web: <u>www.enghouseinteractive.com</u>Email: <u>usa.support@enghouse.com</u>

3. Reference Configuration

The configuration used for the compliance testing is shown in **Figure 1**. The detailed administration of basic connectivity between Communication Manager and Application Enablement Services is not the focus of these Application Notes and will not be described.

The devices used in the compliance testing are shown in the table below. In the compliance testing, the agent and supervisor station extensions were monitored by EICC.

Device Type	Device Number/Extension		
Agent stations	65001, 65002		
Supervisor & failure covering station	65000		

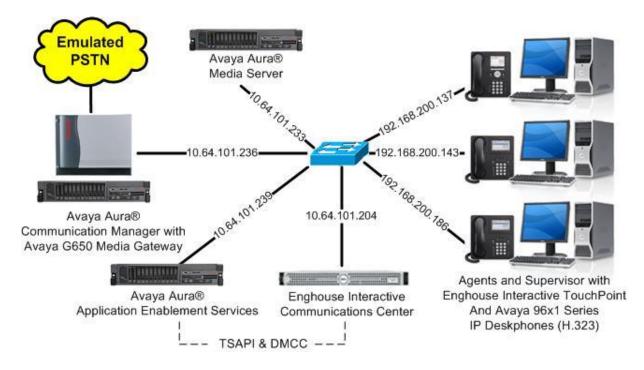


Figure 1: Compliance Testing Configuration

4. Equipment and Software Validated

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

Equipment/Software	Release/Version
Avaya Aura® Communication Manager in Virtual Environment	7.1.1 (7.1.1.0.0.532.23985)
Avaya G650 Media Gateway	NA
Avaya Aura® Media Server in Virtual Environment	7.8.0.333
Avaya Aura® Application Enablement Services in Virtual Environment	7.1.1 (7.1.1.0.0.5-0)
Avaya 9608G & 9641G IP Deskphone (H.323)	6.6506
Enghouse Interactive Communications Center on Windows Server 2012 R2 • Avaya TSAPI Windows Client (csta32.dll) • Avaya DMCC XML	10.0.0.14152 Standard 7.1.1.36 4.2
Enghouse Interactive TouchPoint on Windows 10 Pro	10.0.0.14152

5. Configure Avaya Aura® Communication Manager

This section provides the procedures for configuring Communication Manager. The procedures include the following areas:

- Verify license
- Administer CTI link
- Administer vectors and VDNs
- Administer voicemail coverage path
- Administer agents and supervisors
- Administer virtual IP softphones

5.1. Verify License

Log in to the System Access Terminal to verify that the Communication Manager license has proper permissions for features illustrated in these Application Notes. Use the "display system-parameters customer-options" command to verify that the **Computer Telephony Adjunct Links** customer option is set to "y" on **Page 4**. If this option is not set to "y", then contact the Avaya sales team or business partner for a proper license file.

```
display system-parameters customer-options
                                                                         4 of 12
                                                                  Page
                                OPTIONAL FEATURES
   Abbreviated Dialing Enhanced List? y Audible Message Waiting? y
Access Security Gateway (ASG)? n Authorization Codes? y
                                                   Authorization Codes? y
        Access Security Gateway (ASG)? n
        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? y
                                                         DCS Call Coverage? y
         ASAI Link Plus Capabilities? y DCS with Rerouting? y
```

Navigate to Page 7, and verify that the Vectoring (Basic) customer option is set to "y".

```
7 of 12
display system-parameters customer-options
                                                                    Page
                          CALL CENTER OPTIONAL FEATURES
                           Call Center Release: 7.0
                       BCMS (Basic)? y

Service Level Maximize...
Service Observing (Basic)? y

Service Observing (Basic)? y
                                  ACD? y
                                                                    Reason Codes? y
         BCMS/VuStats Service Level? y
  BSR Local Treatment for IP & ISDN? y
                                           Service Observing (Remote/By FAC)? y
                                            Service Observing (VDNs)? y
                 Business Advocate? n
                    Call Work Codes? y
                                                                      Timed ACW? y
      DTMF Feedback Signals For VRU? y
                                                              Vectoring (Basic)? y
                                                   Vectoring (Prompting)? y
Vectoring (G3V4 Enhanced)? y
                  Dynamic Advocate? n
       Expert Agent Selection (EAS)? y
                             EAS-PHD? y
                                                      Vectoring (3.0 Enhanced)? y
```

5.2. Administer CTI Link

Add a CTI link using the "add cti-link n" command, where "n" is an available CTI link number. Enter an available extension number in the **Extension** field. Note that the CTI link number and extension number may vary. 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

Extension: 60111 Type: ADJ-IP

COR: 1
Name: AES CTI Link

5.3. Administer Vectors and VDNs

Administer a set of vectors and VDNs per EICC installation document [3]. These vectors and VDNs provide general routing and different call treatments to incoming calls. The vectors and VDNs that were used for the compliance testing are shown below.

VDN	Vector	Purpose	
67701	701	Ring treatment	
67702	702	Music treatment	
67703	703	Busy treatment	
67704	704	Failure coverage	
67705	705	Voicemail routing	
67706	700	General routing for the Sales application	
67707	700	General routing for the Support application	
67708	708	Hold treatment	

5.3.1. Failure Coverage

Modify a vector using the "change vector n" command, where "n" is an available vector number. This vector will provide failure coverage and routing to the CTI link defined in **Section 5.2**. Note that the vector **Number** and **route-to number** may vary, and that the **route-to number** is used as the covering point in case of failure from the adjunct routing step.

In the compliance testing, the supervisor extension from **Section 3** was used as the covering point. As shown below, use "SC Fail" as the vector **Name**, with the wait treatment and remaining vector steps as specified in the EICC installation document [3].

```
Change vector 704

CALL VECTOR

Number: 704

Name: SC Fail

Multimedia? n

Basic? y

EAS? y

G3V4 Enhanced? y

ANI/II-Digits? y

ASAI Routing? y

Variables? y

Variables? y

01 adjunct

02 wait-time

03 route-to

04 stop

05
```

Add a VDN using the "add vdn n" command, where "n" is an available extension. Associate this VDN with the newly added vector from above.

• Name: "SC Fail"

• **Destination:** "Vector Number"

• **Vector Number:** The "SC Fail" vector number from above.

add vdn 67704

Page 1 of 3

VECTOR DIRECTORY NUMBER

Extension: 67704

Name*: SC Fail

Destination: Vector Number 704

5.3.2. General Routing

Modify a vector using the "change vector n" command, where "n" is an available vector number. This vector will provide general routing to the CTI link defined in **Section 5.2**. Note that the vector **Number** and **route-to number** may vary, and that the **route-to number** is used as the covering point in case of failure from the adjunct routing step, and set to the failure coverage VDN from **Section 5.3.1**.

Enter a descriptive name for the vector **Name** field, and configure the remaining vector steps as specified in [3].

For each incoming call application, add a VDN using the "add vdn n" command, where "n" is an available extension. Associate this VDN with the newly added vector from above. For the compliance testing, two VDNs were added, as shown below.

Name: A descriptive name.Destination: "Vector Number"

• **Vector Number:** The "EICC User Q" vector number from above.

```
add vdn 67706

Page 1 of 2

VECTOR DIRECTORY NUMBER

Extension: 67706

Name: EICC Sales

Destination: Vector Number 700
```

```
add vdn 67707

VECTOR DIRECTORY NUMBER

Extension: 67707

Name: EICC Support

Destination: Vector Number 700
```

5.3.3. Ring Treatment

Modify a vector using the "change vector n" command, where "n" is an available vector number. This vector will provide ring treatment and routing to the CTI link defined in **Section 5.2**. Note that the vector **Number** and **route-to number** may vary, and that the **route-to number** is used as the covering point in case of failure from the adjunct routing step, and set to the failure coverage VDN from **Section 5.3.1**.

Enter a descriptive name for the vector **Name** field, and configure the remaining vector steps as specified in [3].

```
Change vector 701

Number: 701

Name: SC Ring

Multimedia? n

Basic? y

EAS? y

G3V4 Enhanced? y

ANI/II-Digits? y

Page 1 of 6

CALL VECTOR

Meet-me Conf? n

Lock? n

ASAI Routing? y

Prompting? y

LAI? y

G3V4 Adv Route? y

CINFO? y

BSR? y

Holidays? y

Variables? y

3.0 Enhanced? y

routing link 1

02 wait-time

03 route-to

04 stop

05
```

Add a VDN using the "add vdn n" command, where "n" is an available extension. Associate this VDN with the newly added vector from above.

• Name: "SC Ring"

• **Destination:** "Vector Number"

• **Vector Number:** The "SC Ring" vector number from above.

```
add vdn 67701

Page 1 of 2

VECTOR DIRECTORY NUMBER

Extension: 67701

Name: SC Ring

Destination: Vector Number 701
```

5.3.4. Music Treatment

Modify a vector using the "change vector n" command, where "n" is an available vector number. This vector will provide music treatment and routing to the CTI link defined in **Section 5.2**. Note that the vector **Number** and **route-to number** may vary, and that the **route-to number** is used as the covering point in case of failure from the adjunct routing step, and set to the failure coverage VDN from **Section 5.3.1**.

Enter a descriptive name for the vector **Name** field, and configure the remaining vector steps as specified in [3].

```
Change vector 702

Number: 702

Name: SC Music

Multimedia? n

Basic? y

EAS? y

G3V4 Enhanced? y

ANI/II-Digits? y

Variables? y

Variables? y

1.00

Attendant Vectoring? n

EAS? y

G3V4 Enhanced? y

CINFO? y

BSR? y

Holidays? y

Variables? y

3.0 Enhanced? y

routing link 1

02 wait-time

03 route-to

04 stop

05
```

Add a VDN using the "add vdn n" command, where "n" is an available extension. Associate this VDN with the newly added vector from above.

• Name: "SC Music"

• **Destination:** "Vector Number"

• **Vector Number:** The "SC Music" vector number from above.

add vdn 67702

VECTOR DIRECTORY NUMBER

Extension: 67702

Name: SC Music

Destination: Vector Number 702

5.3.5. Busy Treatment

Modify a vector using the "change vector n" command, where "n" is an available vector number. This vector will provide busy treatment and routing to the CTI link defined in **Section 5.2**. Note that the vector **Number** may vary.

Enter a descriptive name for the vector **Name** field, and configure the remaining vector steps as specified in [3].

```
CALL VECTOR

Number: 703

Name: SC Busy

Multimedia? n

Basic? y

EAS? y

G3V4 Enhanced? y

ANI/II-Digits? y

Prompting? y

LAI? y

G3V4 Adv Route? y

CINFO? y

BSR? y

Holidays? y

Variables? y

3.0 Enhanced? y

or adjunct

routing link 1

or busy

or adjunct

routing link 1
```

Add a VDN using the "add vdn n" command, where "n" is an available extension. Associate this VDN with the newly added vector from above.

• Name: "SC Busy"

• **Destination:** "Vector Number"

• **Vector Number:** The "SC Busy" vector number from above.

add vdn 67703

Page 1 of 2

VECTOR DIRECTORY NUMBER

Extension: 67703

Name: SC Busy
Destination: Vector Number 703

5.3.6. Voicemail Routing

Modify a vector using the "change vector n" command, where "n" is an available vector number. This vector will provide voicemail routing to the CTI link defined in **Section 5.2**. Note that the vector **Number** may vary.

Enter a descriptive name for the vector **Name** field, and configure the remaining vector steps as specified in [3].

```
Change vector 705

Number: 705

Name: SC Voicemail

Multimedia? n

Basic? y

EAS? y

G3V4 Enhanced? y

Prompting? y

Variables? y

Variables? y

1 of 6

CALL VECTOR

Reet-me Conf? n

Lock? n

ASAI Routing? y

ASAI Routing? y

CINFO? y

BSR? y

Holidays? y

Variables? y

1 adjunct

1 couting link 1

1 20 secs hearing ringback

1 of 6

CALL VECTOR
```

Add a VDN using the "add vdn n" command, where "n" is an available extension. Associate this VDN with the newly added vector from above.

Name: "SC Voicemail"Destination: "Vector Number"

• **Vector Number:** The "SC Voicemail" vector number from above.

add vdn 67705

Page 1 of 2

VECTOR DIRECTORY NUMBER

Extension: 67705

Name: SC Voicemail

Destination: Vector Number 705

5.3.7. Hold Treatment

Modify a vector using the "change vector n" command, where "n" is an available vector number. This vector will provide hold treatment and routing to the CTI link defined in **Section 5.2**. Note that the vector **Number** and **route-to number** may vary, and that the **route-to number** is used as the covering point in case of failure from the adjunct routing step, and set to the failure coverage VDN from **Section 5.3.1**.

Enter a descriptive name for the vector **Name** field, and configure the remaining vector steps as specified in [3].

```
Change vector 708

CALL VECTOR

Number: 708

Name: SC Hold

Multimedia? n

Basic? y

EAS? y

G3V4 Enhanced? y

ANI/II-Digits? y

ASAI Routing? y

Prompting? y

LAI? y

G3V4 Adv Route? y

CINFO? y

BSR? y

Holidays? y

Variables? y

3.0 Enhanced? y

1 adjunct

routing link 1

02 wait-time

03 route-to

04 stop

05
```

Add a VDN using the "add vdn n" command, where "n" is an available extension. Associate this VDN with the newly added vector from above.

• Name: "SC Hold"

• Destination: "Vector Number"

• Vector Number: The "SC Hold" vector number from above.

add vdn 67708

VECTOR DIRECTORY NUMBER

Extension: 67708

Name: SC Hold

Destination: Vector Number 708

5.4. Administer Voicemail Coverage Path

Add a coverage path using the "add coverage path n" command, where "n" is an available coverage path number.

For the **Point1** field, enter "v67705" to designate as the first coverage point, where "67705" is the voicemail VDN extension from **Section 5.3.6**.

add coverage path 7			Page 1 of 1			
	COVERAGE PATH					
3	e Path Number: 7					
Cvg Enabled for VDN Ro	-		ter Coverage? n			
Next	Path Number:	Linkage				
COVERAGE CRITERIA						
Station/Group Status	Inside Call	Outside Call				
Active?	n	n				
Busy?	У	У				
Don't Answer?	У	У	Number of Rings: 2			
All?	n	n				
DND/SAC/Goto Cover?	У	У				
Holiday Coverage?	n	n				
COVERAGE POINTS						
Terminate to Coverage A	ets. with Bridge	d Appearances?	n			
Point1: v67705 Rr	ng: Point2:					
Point3:	Point4:					
Point5:	Point6:					

5.5. Administer Agents and Supervisors

Use the "change station n" command, where "n" is first existing agent station extension from **Section 3**. In the **Coverage Path 1** field, enter the voicemail coverage path number from **Section 5.4**.

```
change station 65001

Extension: 65001

Type: 9611

Port: $00102

Name: CM7 Station 1

Location: 1

Location: 1

Location: 19

Speakerphone: 2-way
Display Language: english
Survivable GK Node Name:
Survivable Trunk Dest? Y

STATION OPTIONS

Location: 1

Speakerphone: 2-way
Display Language: english
Survivable Trunk Dest? Y

IP Video Softphone? n
Short/Prefixed Registration Allowed: default
```

Repeat this section for all agents and supervisors. In the compliance testing, two agents and one supervisor were configured as shown below.

list station 65000 count 3				
STATIONS				
Ext/ Hunt-to	Port/ Type	Name/ Surv GK NN	Room/ Move Data Ext	Cv1/ COR/ Cable/ Cv2 COS TN Jack
65000	s00002 9641	CM Supervisor	no	7 1 1 1
65001	s00102 9611	CM Station 1	no	7 1 1 1
65002	S00118 9641	CM Station 2	no	7 1 1 1

5.6. Administer Virtual IP Softphones

Add a virtual softphone using the "add station n" command, where "n" is an available extension number. Enter the following values for the specified fields, and retain the default values for the remaining fields.

• **Type:** "4624"

Name: A descriptive name.Security Code: A desired value.

• IP SoftPhone: "v"

```
add station 67791
                                                                   Page
                                                                          1 of
                                      STATION
Extension: 67791
                                                                          BCC: 0
                                          Lock Messages? n
                                          Security Code: 123456
     Type: 4624
                                                                           TN: 1
                                       Coverage Path 1:
Coverage Path 2:
     Port: IP
                                                                          COR: 1
     Name: EICC Virtual #1
                                                                         cos: 1
                                       Hunt-to Station:
                                                                       Tests? y
STATION OPTIONS
                                           Time of Day Lock Table:
              Location: Time of Day Lock Table:
Loss Group: 19 Personalized Ringing Pattern: 1
Message Lamp Ext: 67791
               Location:
       Speakerphone: 2-way
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
```

Repeat this section to administer the desired number of virtual IP softphones using sequential extension numbers and same security code value. In the compliance testing, two virtual IP softphones were administered as shown below.

list station 67791 count 2						
		STATION	S			
Ext/ Hunt-to	Port/ Type	Name/ Surv GK NN	Move	Room/ Data Ext	Cv1/ C	COR/ Cable/ COS TN Jack
67791	S00027 4624	EICC Virtual #1	no		1	l l 1
67792	s00030 4624	EICC Virtual #2	no		1 1	

6. Configure Avaya Aura® Application Enablement Services

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

- Launch OAM interface
- Verify license
- Administer TSAPI link
- Administer H.323 gatekeeper
- Administer EICC user
- Administer security database
- Administer ports
- Administer TCP settings
- Restart services
- Obtain Tlink name

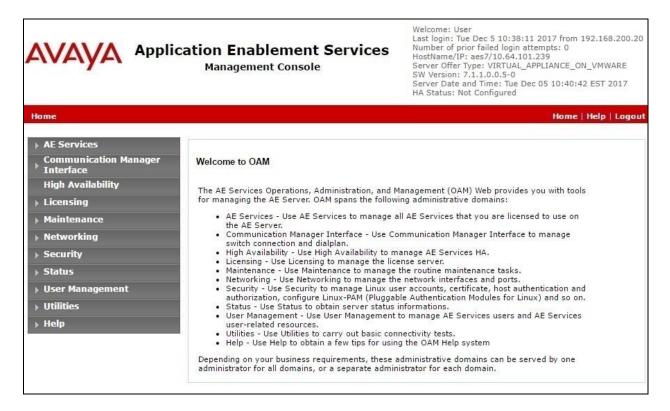
6.1. Launch OAM Interface

Access the OAM web-based interface by using the URL "https://ip-address" in an Internet browser window, where "ip-address" is the IP address of the Application Enablement Services server.

The **Please login here** screen is displayed. Log in using the appropriate credentials.



The **Welcome to OAM** screen is displayed next.

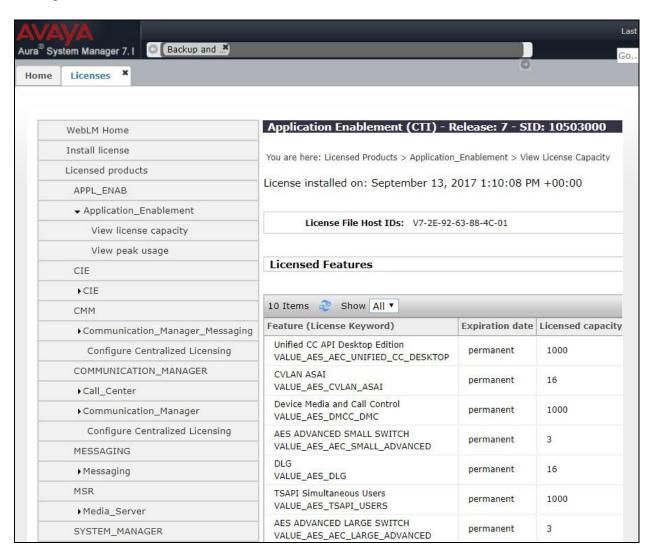


6.2. Verify License



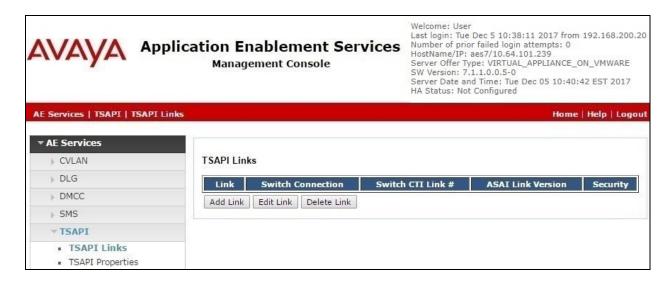
Select Licensed products \rightarrow APPL_ENAB \rightarrow Application_Enablement in the left pane, to display the Application Enablement (CTI) screen in the right pane.

Verify that there are sufficient licenses for **TSAPI Simultaneous Users** and **Device Media and Call Control**, as shown below. The TSAPI license is used for device monitoring and the DMCC license is used for the virtual IP softphones. Also verify that there is an applicable advanced switch license, in this case **AES ADVANCED LARGE SWITCH**, which is needed for adjunct routing.



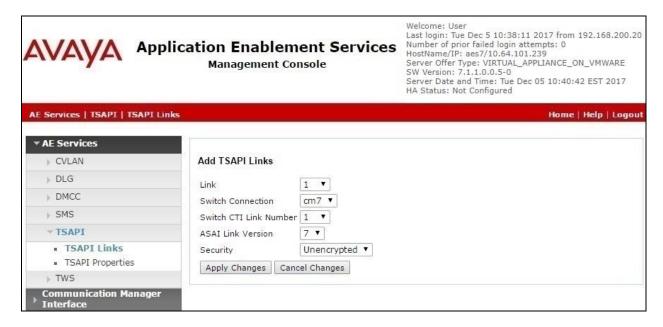
6.3. Administer TSAPI Link

Select **AE Services** → **TSAPI** → **TSAPI Links** from the left pane of the **Management Console**, to administer a TSAPI link. The **TSAPI Links** screen is displayed, as shown below. Click **Add Link**.



The **Add TSAPI Links** screen is displayed next.

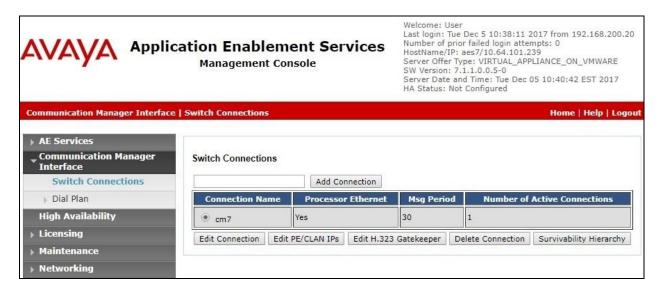
The **Link** field is only local to the Application Enablement Services server, and may be set to any available number. For **Switch Connection**, select the relevant switch connection from the drop-down list. In this case, the existing switch connection "cm7" is selected. For **Switch CTI Link Number**, select the CTI link number from **Section 5.2**. Retain the default values in the remaining fields.



6.4. Administer H.323 Gatekeeper

Select Communication Manager Interface → Switch Connections from the left pane. The Switch Connections screen shows a listing of the existing switch connections.

Locate the connection name associated with the relevant Communication Manager, in this case "cm7", and select the corresponding radio button. Click **Edit H.323 Gatekeeper**.



The **Edit H.323 Gatekeeper** screen is displayed next. Enter the IP address of a C-LAN circuit pack or the Processor C-LAN on Communication Manager to use as the H.323 gatekeeper, in this case "10.64.101.236" as shown below. Click **Add Name or IP**.



6.5. Administer EICC User

Select User Management \rightarrow User Admin \rightarrow Add User from the left pane, to display the Add User screen in the right pane.

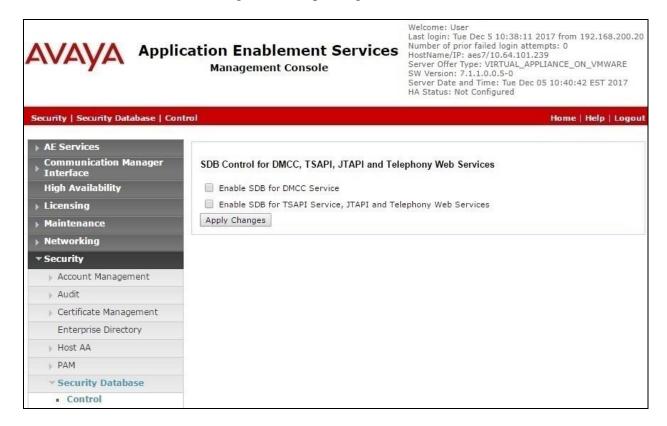
Enter desired values for **User Id**, **Common Name**, **Surname**, **User Password**, and **Confirm Password**. For **CT User**, select "Yes" from the drop-down list. Retain the default value in the remaining fields.



6.6. Administer Security Database

Select Security \rightarrow Security Database \rightarrow Control from the left pane, to display the SDB Control for DMCC, TSAPI, JTAPI and Telephony Web Services screen in the right pane. Uncheck both fields below.

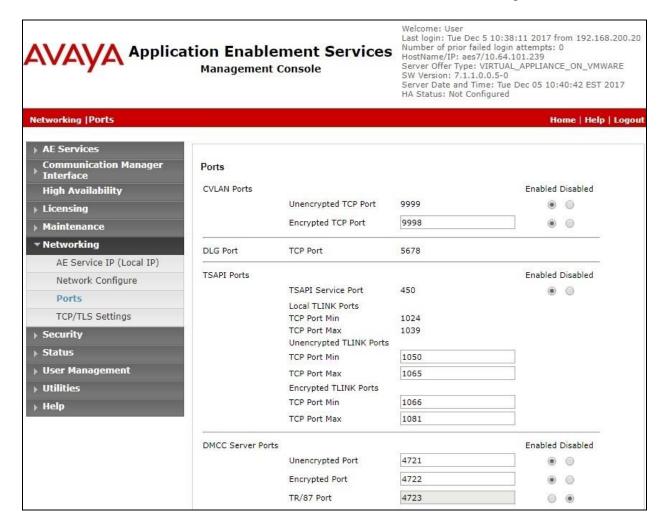
In the event that the security database is used by the customer with parameters already enabled, then follow reference [2] to configure access privileges for the EICC user from **Section 6.5**.



6.7. Administer Ports

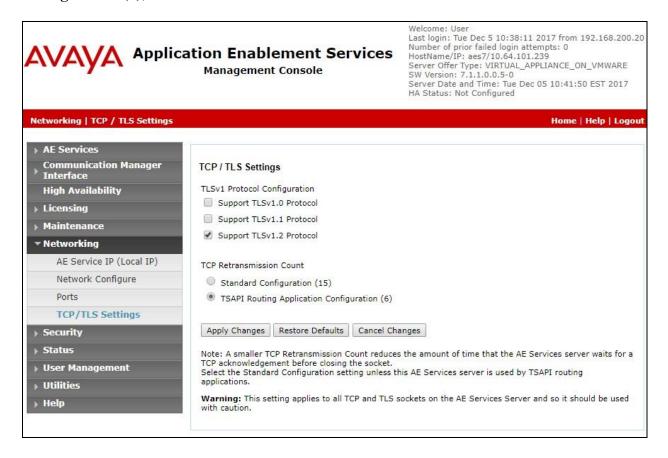
Select **Networking** \rightarrow **Ports** from the left pane, to display the **Ports** screen in the right pane.

In the **DMCC Server Ports** section, select the radio button for **Unencrypted Port** under the **Enabled** column, as shown below. Retain the default values in the remaining fields.



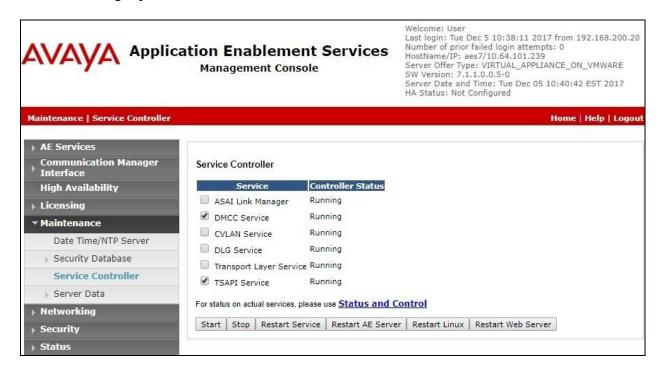
6.8. Administer TCP Settings

Select Networking \rightarrow TCP/TLS Settings from the left pane, to display the TCP/TLS Settings screen in the right pane. For TCP Retransmission Count, select TSAPI Routing Application Configuration (6), as shown below.



6.9. Restart Services

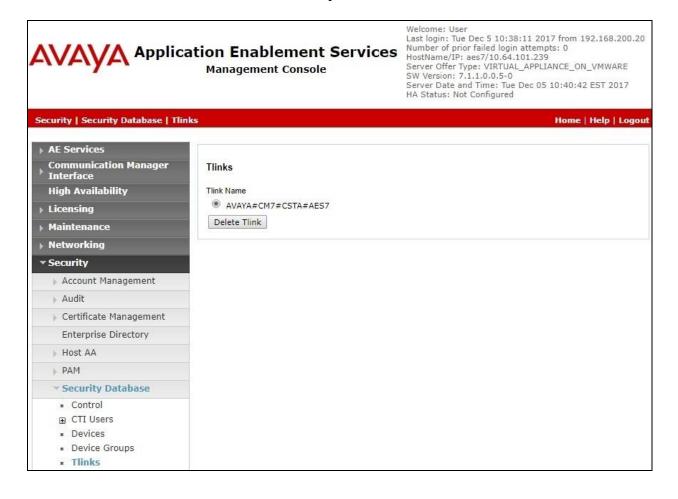
Select Maintenance \rightarrow Service Controller from the left pane, to display the Service Controller screen in the right pane. Check DMCC Service and TSAPI Service, and click Restart Service.



6.10. Obtain Tlink Name

Select Security Security Database Tlinks from the left pane. The Tlinks screen shows a listing of the Tlink names. A new Tlink name is automatically generated for the TSAPI service. Locate the Tlink name associated with the relevant switch connection, which would use the name of the switch connection as part of the Tlink name. Make a note of the associated Tlink name, to be used later for configuring EICC.

In this case, the associated Tlink name is "AVAYA#CM7#CSTA#AES7". Note the use of the switch connection "CM7" from **Section 6.3** as part of the Tlink name.



7. Configure Enghouse Interactive Communications Center

This section provides the procedures for configuring the EICC server. The procedures include the following areas:

- Administer phone system type
- Administer phone system data
- Administer queues
- Administer agent login class
- Administer agents and supervisors
- Administer mailboxes
- Administer lines

The configuration of EICC is typically performed by Enghouse Interactive installation technicians or third party resellers. The procedural steps are presented in these Application Notes for informational purposes.

Prior to configuration, the relevant Avaya TSAPI client is assumed to be installed on the EICC server, and that the TSAPI client has been configured with the IP address of the Application Enablement Services server as part of installation.

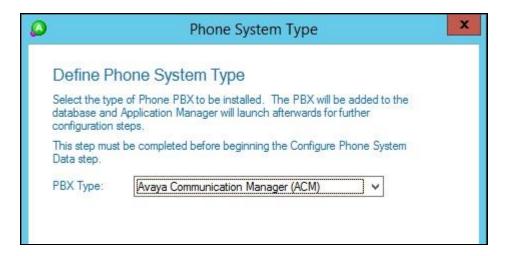
7.1. Administer Phone System Type

At the conclusion of installation, the **Installation Process** screen will be displayed by the Installation Wizard. Follow [3] to import licenses and configure the default company.

The **Installation Process** screen shown below is displayed next. Click the **Run** icon associated with **Define Phone System Type**.

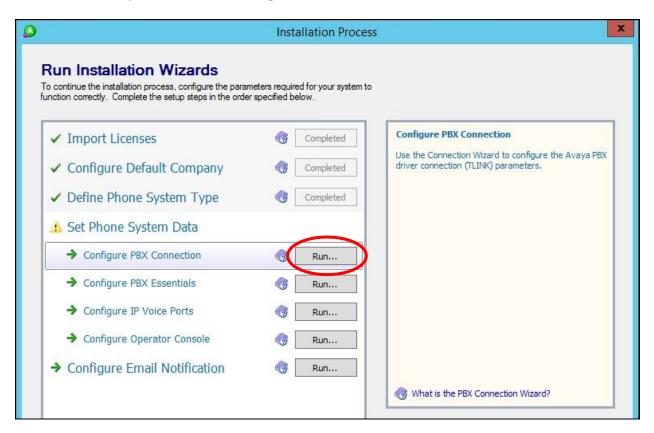


The **Phone System Type** screen is displayed. For **PBX Type**, select "Avaya Communication Manager (ACM)".

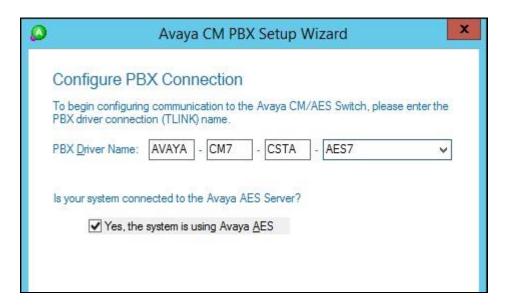


7.2. Administer Phone System Data

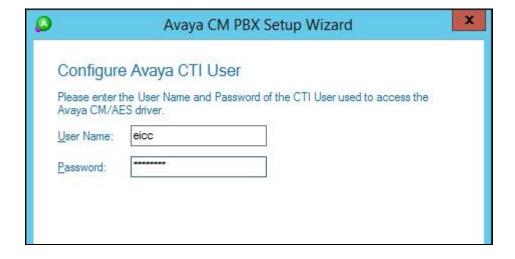
The **Installation Process** screen shown below is displayed next. Click the **Run** icon associated with **Set Phone System Data** → **Configure PBX Connection**.



The Avaya CM PBX Setup Wizard → Configure PBX Connection screen is displayed. For PBX Driver Name, enter the Tlink name from Section 6.10 as shown below. Retain the default value in the remaining field.



The Avaya CM PBX Setup Wizard → Configure Avaya CTI User screen is displayed next. Enter the EICC user credentials from Section 6.5.



The Avaya CM PBX Setup Wizard → Configure ACM Soft Ports screen is displayed. Enter the following values for the specified fields.

• ACM Switch Connection Name: The relevant switch connection name from Section 6.3.

• **ACM IP Address:** IP address of H.323 gatekeeper from **Section 6.4**.

• **AES IP Address:** IP address of Application Enablement Services server.

• **DMCC TCP Port:** "4721"

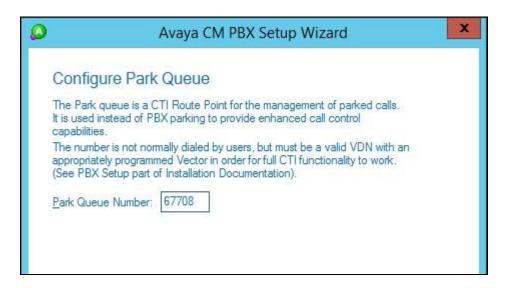
DMCC User:

 DMCC Password:
 Global SoftPort Password:

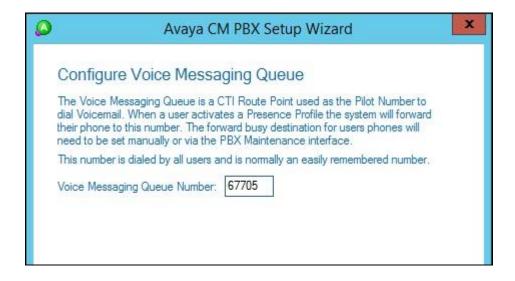
 The EICC user credentials from Section 6.5.
 The security code value from Section 5.6.



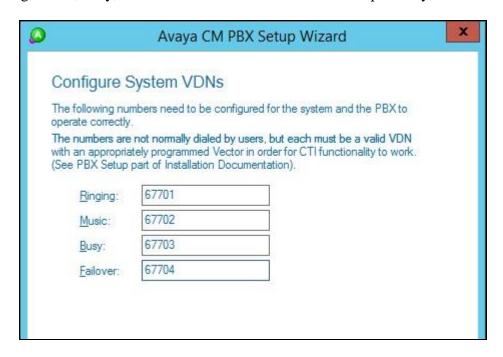
Continue with the Installation Wizard until the **Avaya CM PBX Setup Wizard** → **Configure Park Queue** screen is displayed. For **Park Queue Number**, enter the extension of the hold VDN from **Section 5.3.7**.



The Avaya CM PBX Setup Wizard → Configure Voice Messaging Queue screen is displayed next. For Voice Messaging Queue Number, enter the extension of the voicemail VDN from Section 5.3.6.

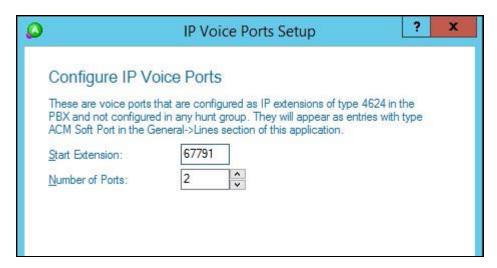


The Avaya CM PBX Setup Wizard → Configure System VDNs screen is displayed next. Enter the ring, music, busy, and failure VDNs from Section 5.3 respectively, as shown below.



Continue with the Installation Wizard until the IP Voice Ports Setup \rightarrow Configure IP Voice Ports screen is displayed. For Start Extension, enter the first virtual IP softphone extension from Section 5.6. For Number of Ports, select the total number of virtual IP softphones from Section 5.6, in this case "2".

Follow [3] to complete the Installation Wizard and subsequent CTI server setup via Application Manager.

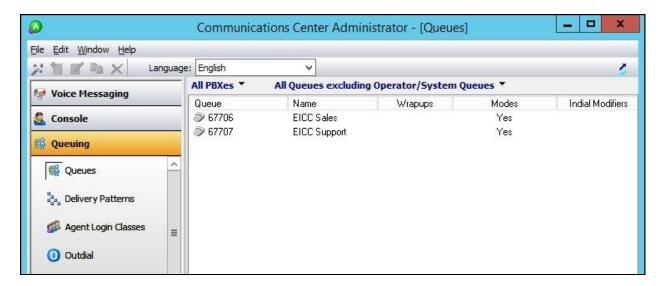


7.3. Administer Queues

The **Administrator** screen is displayed upon completion of the Installation Wizard and CTI server setup. Select **Queuing \rightarrow Queues** from the left pane, followed by the **Add Wizard** icon located at the upper left of the screen.

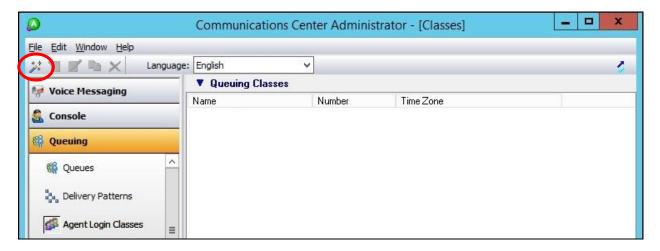


Follow the **Adding a New Queue Wizard** in the subsequent screens (not shown) to configure a new queue for each general routing VDN in **Section 5.3.2**. In the compliance testing, two queues were created as shown below.

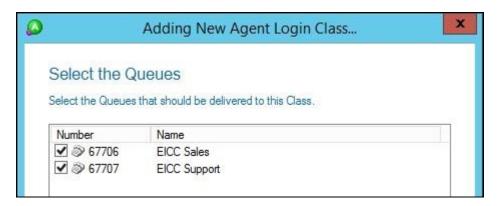


7.4. Administer Agent Login Class

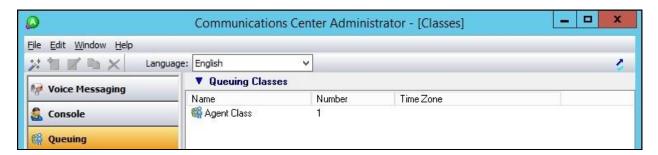
Select Queuing \rightarrow Agent login Classes from the left pane, followed by the Add Wizard icon located at the upper left corner of the screen.



Follow the **Adding New Agent Login Class Wizard** in the subsequent screens to configure a new agent login class. In the **Select the Queues** screen, select the queues created from **Section 7.3**, as shown below.



In the compliance testing, one agent login class was created.



7.5. Administer Agents and Supervisors

Select **Queuing** \rightarrow **Agents** from the left pane, followed by the **Add Wizard** icon located at the upper left corner of the screen.



Follow the **Add Agent Wizard** in the subsequent screens to configure a corresponding entry for each agent and supervisor in **Section 3**. In the **Select Agent Login Class** screen, select the agent login class created from **Section 7.4**, as shown below.

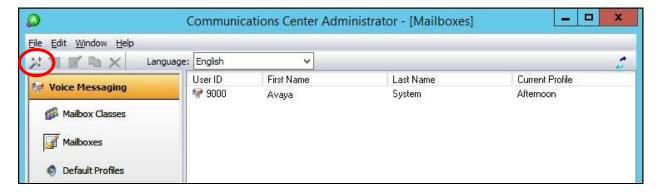


In the compliance testing, two agents and one supervisor were created.

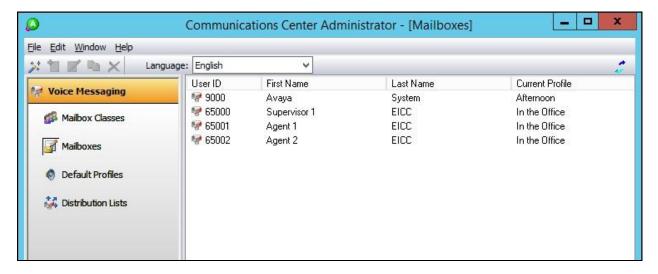


7.6. Administer Mailboxes

Select Voice Messaging → Mailboxes from the left pane, followed by the Add Wizard icon located at the upper left corner of the screen.



Follow the **Add Mailboxes Wizard** in the subsequent screens (not shown) to configure a corresponding mailbox for each agent and supervisor from **Section 7.5**. In the compliance testing, three mailboxes were created.

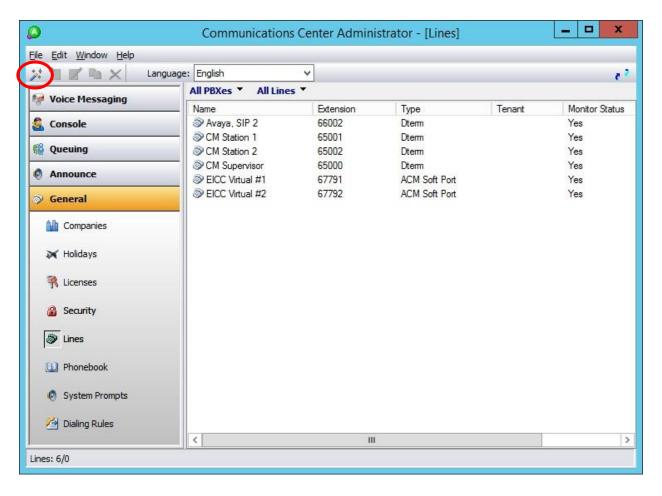


7.7. Administer Lines

Select General → Lines from the left pane, followed by the Add Wizard icon located at the upper left corner of the screen. Follow the Adding Line Wizard in the subsequent screens (not shown) to configure a corresponding line for each agent and supervisor from Section 7.5.

Note that the lines for virtual IP softphones were created automatically, and that lines for agents and supervisors can either be created manually using the wizard, or by having each agent and supervisor dial a monitored VDN for EICC to "learn" the extension and create the line automatically.

In the compliance testing, all lines were created automatically with agents and supervisor dialing the voicemail VDN for EICC to "learn" the extensions.



8. Verification Steps

This section provides the tests that can be performed to verify proper configuration of Communication Manager, Application Enablement Services, and EICC.

8.1. Verify Avaya Aura® Communication Manager

On Communication Manager, verify status of the administered CTI link by using the "status aesvcs cti-link" command. Verify that the **Service State** is "established" for the CTI link number administered in **Section 5.2**, as shown below.

```
status aesvcs cti-link

AE SERVICES CTI LINK STATUS

CTI Version Mnt AE Services Service Msgs Msgs
Link Busy Server State Sent Rcvd

1 7 no aes7 established 26 14
```

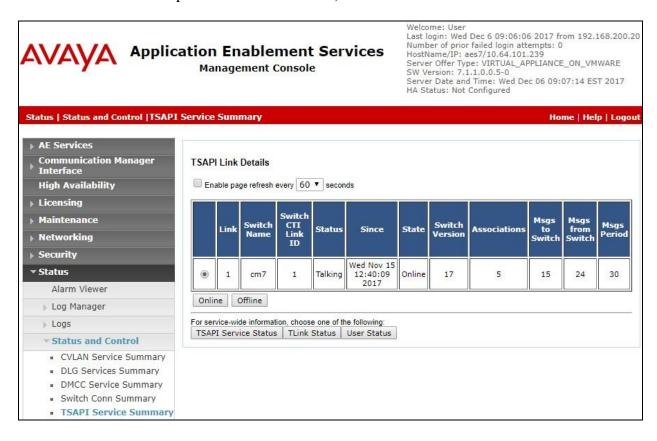
Verify the registration status of virtual IP softphones by using the "list registered-ip-stations" command. Verify that all virtual IP softphone from **Section 5.6** are displayed along with the IP address of the Application Enablement Services server, as shown below.

list registered-ip-stations				
REGISTERED			ERED	IP STATIONS
				Station IP Address/ Gatekeeper IP Address
65000	9641	IP_Phone 6.6506	tls	192.168.200.186 10.64.101.236
65001	9611		tls	192.168.200.137 10.64.101.236
65002	9641		tls	192.168.200.143 10.64.101.236
67791	4624 1	IP_API_A 3.2040	tcp	10.64.101.239 10.64.101.236
67792	4624 1	IP_API_A 3.2040	tcp	10.64.101.239 10.64.101.236

8.2. Verify Avaya Aura® Application Enablement Services

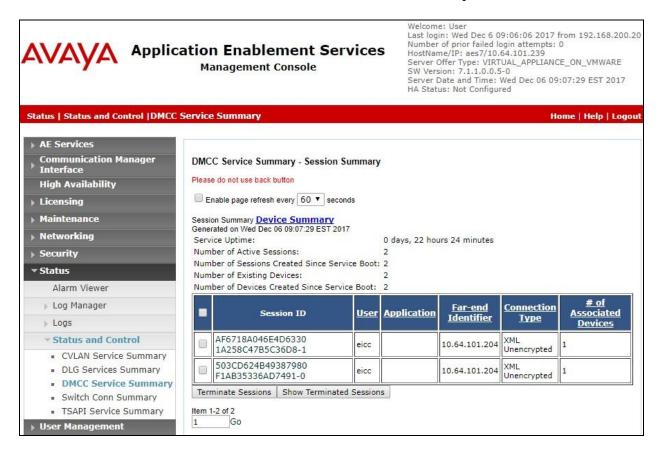
On Application Enablement Services, verify the status of the TSAPI link by selecting **Status Status and Control TSAPI Service Summary** from the left pane. The **TSAPI Link Details** screen is displayed.

Verify the **Status** is "Talking" for the TSAPI link administered in **Section 6.3**, and that the **Associations** column reflects the total number of agents and supervisor from **Section 3** plus the number of virtual IP softphones from **Section 5.6**, in this case "5".



Verify the status of the DMCC link by selecting **Status** → **Status** and **Control** → **DMCC Service Summary** from the left pane. The **DMCC Service Summary** – **Session Summary** screen is displayed.

Verify the **User** column shows action sessions with the EICC user name from **Section 6.5**, and that the total number of sessions reflects the number of virtual IP softphones from **Section 5.6**.



8.3. Verify Enghouse Interactive Communications Center

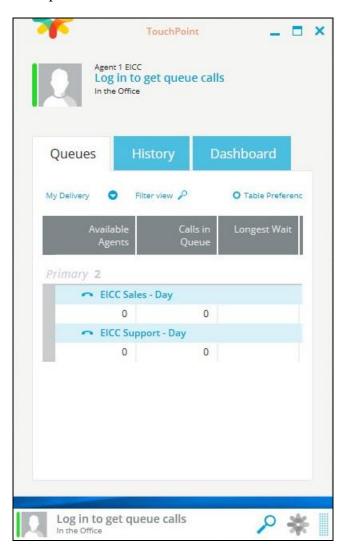
From the agent desktop, double-click on the **TouchPoint** shortcut icon shown below, which was created as part of TouchPoint installation.



The **Enghouse Interactive TouchPoint** login screen below is displayed. Enter the login name associated with an agent from **Section 7.5**, and use the generic default PIN value from EICC. Retain the default value in the remaining field.



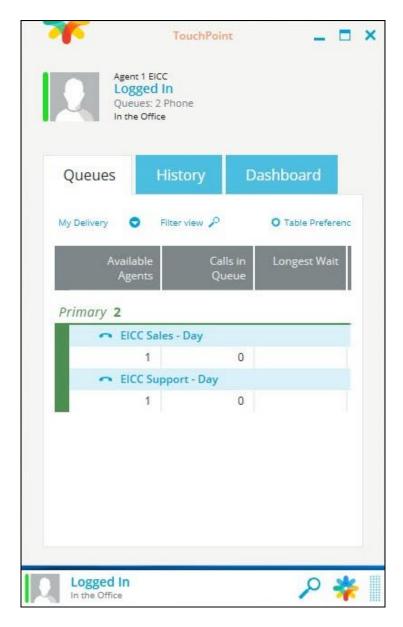
The main **TouchPoint** screen, also referred to as the Statistics Window is displayed, along with a Call Bar above the system tray, as shown below. From the Statistics Window, click on **Log in to get queue calls** toward the top of the screen.



In the updated Statistics Window shown below, select **Log in to Queues**.

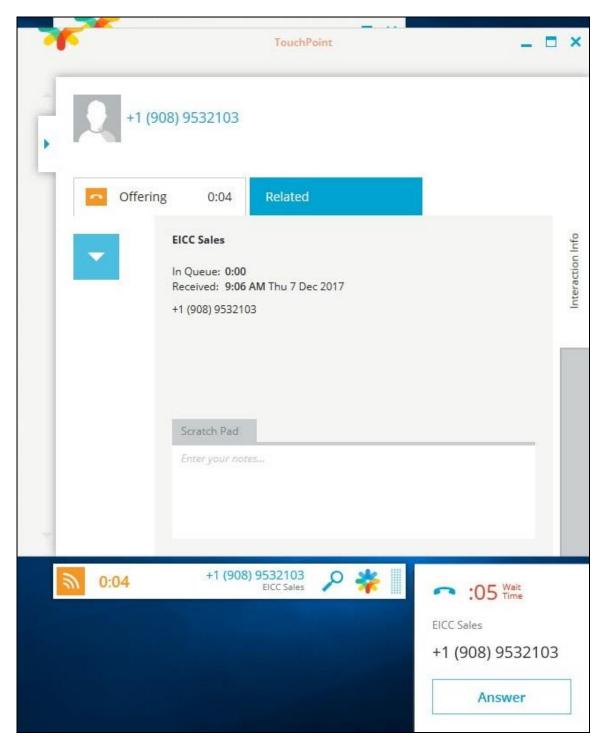


Verify that both the Statistics Window and Call Bar are updated to reflect **Logged In**, as shown below.



Make an incoming call from PSTN to a general routing VDN in **Section 5.3.2**. Verify that the agent desktop is populated with an **Interaction Info** screen with an **Offering** tab, along with a Pop-up Notification box, and that the Call Bar is updated to reflect the active call.

Click **Answer** in the Pop-up Notification box, and verify that the agent is connected to the PSTN caller with two-way talk paths.



9. Conclusion

These Application Notes describe the configuration steps required for Enghouse Interactive Communications Center 10.0 to successfully interoperate with Avaya Aura® Communication Manager 7.1 using Avaya Aura® Application Enablement Services 7.1. All feature and serviceability test cases were completed with observations noted in **Section 2.2**.

10. Additional References

This section references the product documentation relevant to these Application Notes.

- **1.** Administering Avaya Aura® Communication Manager, Release 7.1.1, Issue 2, August 2017, available at http://support.avaya.com.
- **2.** Administering and Maintaining Aura® Application Enablement Services, Release 7.1.1, Issue 3, September 2017, available at http://support.avaya.com.
- **3.** *CC 10.0 First-time Installation and Server Setup Avaya Communication Manager*, June 2017, available at https://partnerportal.enghouseinteractive.com/Sys/Document/index.
- **4.** *CC 10.0 for PBX Programming Manual Avaya Communication Manager*, June 2017, available at https://partnerportal.enghouseinteractive.com/Sys/Document/index.

©2018 Avaya Inc. All Rights Reserved.

Avaya and the Avaya Logo are trademarks of Avaya Inc. All trademarks identified by ® and TM are registered trademarks or trademarks, respectively, of Avaya Inc. All other trademarks are the property of their respective owners. The information provided in these Application Notes is subject to change without notice. The configurations, technical data, and recommendations provided in these Application Notes are believed to be accurate and dependable, but are presented without express or implied warranty. Users are responsible for their application of any products specified in these Application Notes.

Please e-mail any questions or comments pertaining to these Application Notes along with the full title name and filename, located in the lower right corner, directly to the Avaya DevConnect Program at devconnect@avaya.com.