

Avaya Solution & Interoperability Test Lab

Application Notes for Zeacom Communications Center with Avaya Aura® Communication Manager using Avaya Aura® Application Enablement Services 6.3 – Issue 1.0

Abstract

These Application Notes describe the configuration steps required for Zeacom Communications Center to interoperate with Avaya Aura® Communication Manager using Avaya Aura® Application Enablement Services 6.3. Zeacom 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.

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 Zeacom Communications Center to interoperate with Avaya Aura® Communication Manager using Avaya Aura® Application Enablement Services 6.3. Zeacom 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 (TSAPI) and Device, Media, and Call Control (DMCC) interface.

The TSAPI interface is used by Zeacom Communications Center to query and monitor devices such as call answering station user extensions on Avaya Aura® Communication Manager. Incoming calls are routed by Zeacom Communications Center using the TSAPI adjunct routing capability.

The call answering station users (referred to as agents) have desktop computers running the Zeacom Executive Desktop application. Call related actions such as answering of incoming calls can be initiated via the agent telephone, or via the agent desktop by using the TSAPI call control capabilities. The Zeacom Communications Center server populates the answering agent's desktop screen with call related information, by using the received TSAPI event reports for the monitored devices.

The DMCC interface is used by Zeacom Communications Center to support the voicemail application via virtual IP softphones. Calls to the Voicemail VDN are routed by Zeacom Communications Center over an available virtual IP softphone. The TSAPI set value capability is used to support enable/disable of call forwarding and message waiting lamps.

2. General Test Approach and Test Results

The feature test cases were performed both automatically and manually. Upon start of the Communications Center application, the application automatically uses TSAPI to query device name, requests device monitoring, and registers for VDN routing. The application also automatically uses DMCC to register the virtual IP softphones.

For the manual part of the testing, incoming calls were made to the general routing VDNs. The Communications Center server used the query results and event reports to track agent states, and specified calls to be routed to available agents. Manual call controls from both the agent telephones and the agent desktop computers were exercised to verify call control features such as answering and transferring of calls.

Voicemail was tested by not answering the calls at the agents, and have the calls cover to the Zeacom Voicemail application with proper activation of the agent message waiting lamps. Manual calls were then made from the agents to the Voicemail VDN to retrieve the voice messages and verify proper deactivation of the message waiting lamps.

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

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 Communications Center 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.

2.1. Interoperability Compliance Testing

The interoperability compliance test included feature and serviceability testing.

The feature testing focused on verifying the following on Communications Center:

- Use of TSAPI query service to query device names.
- Use of TSAPI event report service to monitor agent and virtual IP softphone extensions.
- Use of TSAPI routing service to route incoming calls.
- Use of TSAPI set value service to activate/deactivate call forwarding and message waiting indicator.
- Use of TSAPI call control service to support manual call control actions initiated from the agent desktop and to automatically handle inbound calls to the virtual IP softphones.
- Use of DMCC registration service to register and un-register the virtual IP softphones.
- Proper handling of call scenarios involving inbound, outbound, ACD, non-ACD, drop, hold/reconnect, voicemail, transfer, conference, call forwarding, and supervisor monitor.

The serviceability testing focused on verifying the ability of Communications Center to recover from adverse conditions, such as disconnecting/reconnecting the Ethernet connection to Communications Center.

2.2. Test Results

All test cases were executed and verified. The one observation from the compliance testing is Communications Center creates one DMCC session per virtual IP softphone.

2.3. Support

Technical support on Communications Center can be obtained through the following:

Phone: (800) 513-9002Web: www.zeacom.com

• Email: <u>usasupport@zeacom.com</u>

3. Reference Configuration

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 contact center devices used in the compliance testing are shown in the table below. In the compliance testing, Communications Center monitored two agents and one supervisor station extensions shown below.

Device Type	Device Number/Extension		
VDNs	45901-45908		
Vectors	900-905, 908		
Agent stations	45001-45002		
Supervisor & Failure covering station	45000		

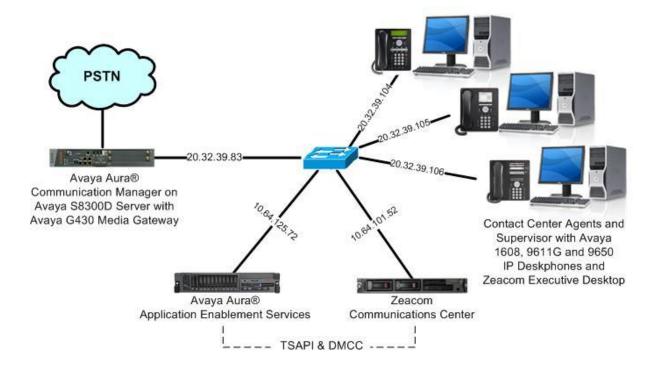


Figure 1: Compliance Testing Configuration

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 on Avaya S8300D Server with Avaya G430 Media Gateway	6.3.2 (R016x.03.0.124.0-21053)
Avaya Aura® Application Enablement Services	6.3.1 (6.3.1.0.19-0)
Avaya 1608 IP Deskphone (H.323)	1.340B
Avaya 9611G IP Deskphone (H.323)	6.3037
Avaya 9650 IP Deskphone (H.323)	3.210A
Zeacom Communications Center on Windows Server 2008 R2 Enterprise • AnnouncePort.exe • Avaya TSAPI Windows Client • Avaya DMCC XML	7.0.0.1288 SP1 7.0.0.1400 6.3.1.502 6.2
Zeacom Executive Desktop	7.0.0.1288

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 3**. If this option is not set to "y", then contact the Avaya sales team or business partner for a proper license file.

```
3 of 11
display system-parameters customer-options
                                                                      Page
                                  OPTIONAL FEATURES
    Abbreviated Dialing Enhanced List? y Audible Message Waiting? y
Access Security Gateway (ASG)? n Authorization Codes? y
Apalog Trunk Incoming Call ID? 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
```

Navigate to Page 6, and verify that the Vectoring (Basic) customer option is set to "y".

```
display system-parameters customer-options
                                                                  Page 6 of 11
                          CALL CENTER OPTIONAL FEATURES
                           Call Center Release: 6.0
                                 ACD? y
                                                                  Reason Codes? y
        BCMS (Basic)? y Service Level Maximizer? n
BCMS/VuStats Service Level? y Service Observing (Basic)? 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,
Vectoring (G3V4 Enhanced)? y
                  Dynamic Advocate? n
      Expert Agent Selection (EAS)? y
                            EAS-PHD? 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: 40001 Type: ADJ-IP

COR: 1
Name: AES CTI Link

5.3. Administer Vectors and VDNs

Administer a set of vectors and VDNs per Communications Center 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
49101	101	Ring treatment
49102	102	Music treatment
49103	103	Busy treatment
49104	104	Failure coverage
49105	105	Voicemail routing
49106	100	General routing for the Sales application
49107	100	General routing for the Support application
49108	108	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 Communications Center installation document [3].

```
Change vector 104

CALL VECTOR

Number: 104

Name: SC Fail

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

701 adjunct

102 wait-time

103 route-to

104 stop

105
```

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 49104

Page 1 of 3

VECTOR DIRECTORY NUMBER

Extension: 49104

Name*: SC Fail

Destination: Vector Number 104

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 "Zeacom User Q" vector number from above.

add vdn 49106	VECTOR DIRECTORY NUMBER	Page	1 of	2
	Extension: 49106 Name: Zeacom Sales Destination: Vector Number	100		

```
add vdn 49107

VECTOR DIRECTORY NUMBER

Extension: 49107

Name: Zeacom Support

Destination: Vector Number 100
```

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 101

CALL VECTOR

Number: 101

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

Touting link 1

02 wait-time

60 secs hearing ringback

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 49101

VECTOR DIRECTORY NUMBER

Extension: 49101

Name: SC Ring

Destination: Vector Number 101
```

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 102

CALL VECTOR

Number: 102

Name: SC Music

Multimedia? n

Basic? y

EAS? y

G3V4 Enhanced? y

ANI/II-Digits? y

Variables? y

Variables? y

1.06

CALL VECTOR

Page 1 of 6

CALL VECTOR

Number: 102

Name: SC Music

Lock? n

Lock? n

ASAI Routing? 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 49102

VECTOR DIRECTORY NUMBER

Extension: 49102

Name: SC Music

Destination: Vector Number 102

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].

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 49103

VECTOR DIRECTORY NUMBER

Extension: 49103

Name: SC Busy
Destination: Vector Number 103

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].

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: "Voicemail"Destination: "Vector Number"

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

add vdn 49105

VECTOR DIRECTORY NUMBER

Extension: 49105

Name: Voicemail

Destination: Vector Number 105

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 108

CALL VECTOR

Number: 108

Name: SC Hold

Multimedia? n

Basic? y

EAS? y

G3V4 Enhanced? y

ANI/II-Digits? y

Page 1 of 6

CALL VECTOR

Met-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 Hold"

• Destination: "Vector Number"

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

add vdn 49108

VECTOR DIRECTORY NUMBER

Extension: 49108

Name: SC Hold

Destination: Vector Number 108

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 "v49105" to designate as the first coverage point, where "49105" 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 R	-		ter Coverage? n
Nex	t 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	Pts. with Bridge	d Appearances? r	1
Point1: v49105 R	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 45001
                                                                                     Page 1 of 5
                                              STATION
      Asion: 45001 Lock Messages? n BCC: 0
Type: 1608 Security Code: * TN: 1
Port: S00000 Coverage Path 1: 7 COR: 1
Name: G430 Station 1 Coverage Path 2: COS: 1
Hunt-to Station: Tests? y
Extension: 45001
STATION OPTIONS
                                                        Time of Day Lock Table:
         Loss Group: 19

Personalized Ringing Pattern: 1

Message Lamp Ext: 45

Speakerphone: 2-way

Display Language: english
able GK Node Name:
                                                              Message Lamp Ext: 45001
 Survivable GK Node Name:
      Survivable COR: internal
                                                             Media Complex Ext:
   Survivable Trunk Dest? y
                                                                     IP SoftPhone? n
                                                              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	45000 cc	ount 3					
		STATIC	NS				
Ext/ Hunt-to	Port/ Type	Name/ Surv GK NN	Move	Room/ Data Ext	Cv1/ Cv2		R/ Cable/ OS TN Jack
45000	s00033 9650	G430 Station 0	no		7	1 1	1
45001	\$00000 1608	G430 Station 1	no		7	1 1	1
45002	s00008 9611	G430 Station 2	no		7	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 45991
                                                              Page
                                                                    1 of
                                    STATION
Extension: 45991
                                                                    BCC: 0
                                       Lock Messages? n
                                       Security Code: 123456
    Type: 4624
                                                                     TN: 1
                                     Coverage Path 1:
Coverage Path 2:
    Port: S00036
                                                                    COR: 1
    Name: Zeacom Virtual #1
                                                                    cos: 1
                                     Hunt-to Station:
                                                                  Tests? y
STATION OPTIONS
                                         Time of Day Lock Table:
             Loss Group: 19 Personalized Ringing Pattern: 1
                                             Message Lamp Ext: 45991
       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 4	15991 co	unt 2				
		STATIONS				
Ext/ Hunt-to	Port/ Type	Name/ Surv GK NN	Move	Room/ Data Ext	Cv1/ C Cv2	COR/ Cable/ COS TN Jack
45991	s00036 4624	Zeacom Virtual #1	no		1 1	. 1
45992	\$00039 4624	Zeacom Virtual #2	no		1 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
- Disable security database
- Administer TCP Settings
- Restart services
- Obtain Tlink name
- Administer Zeacom user
- Enable ports

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.

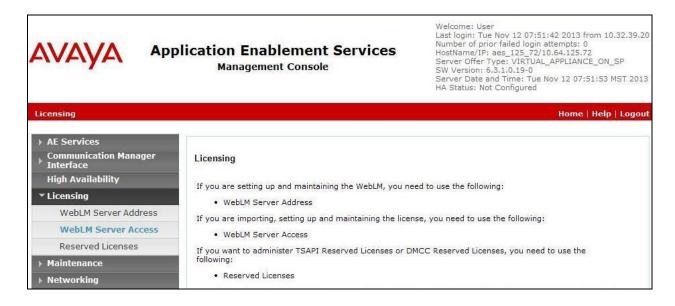


The **Welcome to OAM** screen is displayed next.



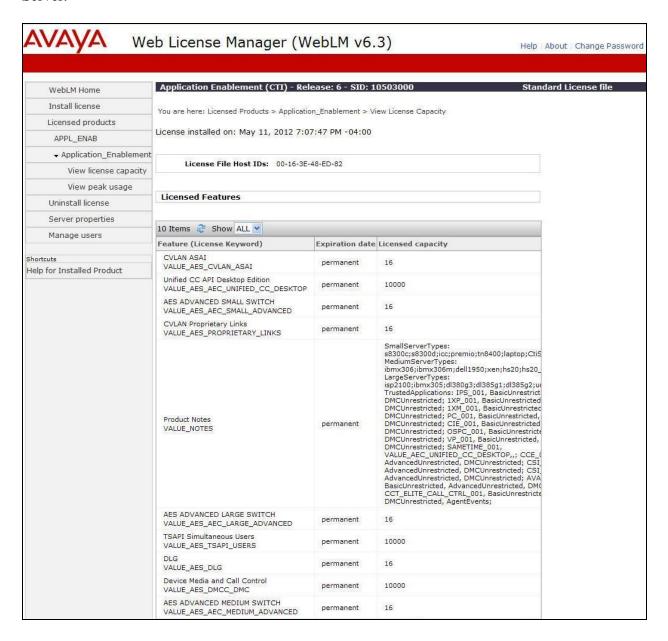
6.2. Verify License

Select Licensing \rightarrow WebLM Server Access in the left pane, to display the Web License Manager pop-up screen (not shown), and log in using the appropriate credentials.



The Web License Manager screen below is displayed. 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. Note that 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 SMALL SWITCH** for the S8300D Server.



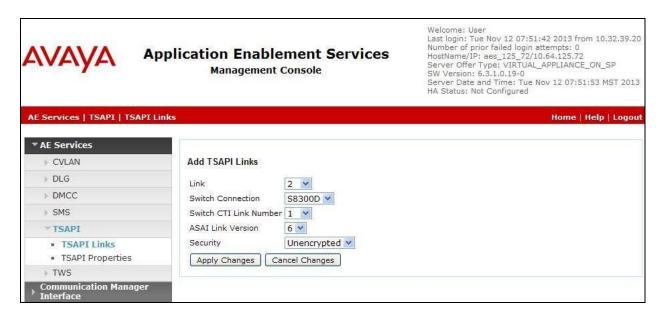
6.3. Administer TSAPI Link

To administer a TSAPI link, select **AE Services** → **TSAPI Links** from the left pane of the **Management Console**. 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 "S8300D" 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 \rightarrow 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 "S8300D", and select the corresponding radio button. Click **Edit H.323 Gatekeeper**.

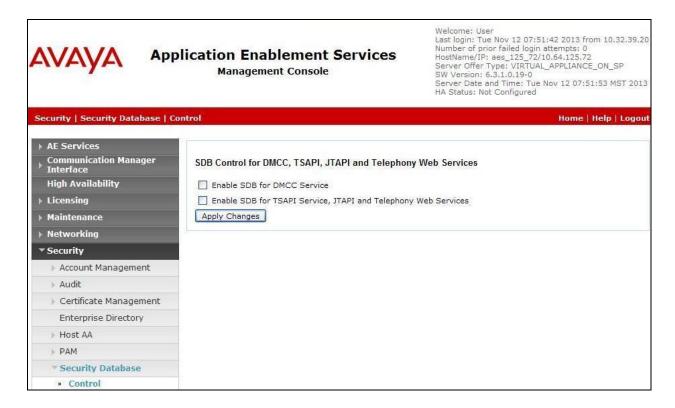


The **Edit H.323 Gatekeeper** screen is displayed. Enter the IP address of a C-LAN circuit pack or the Processor C-LAN on Communication Manager to be used as the H.323 gatekeeper, in this case "10.32.39.83" as shown below. Click **Add Name or IP**.



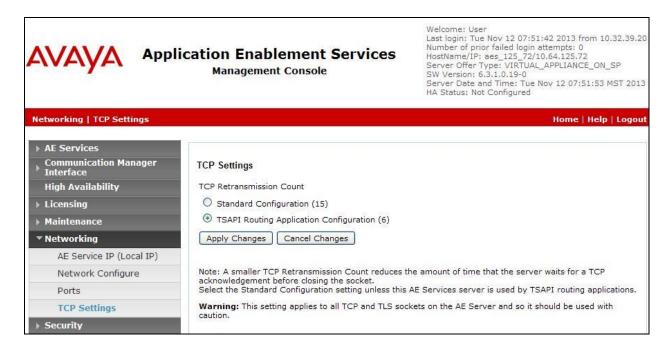
6.5. Disable Security Database

Select Security Security Database 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.



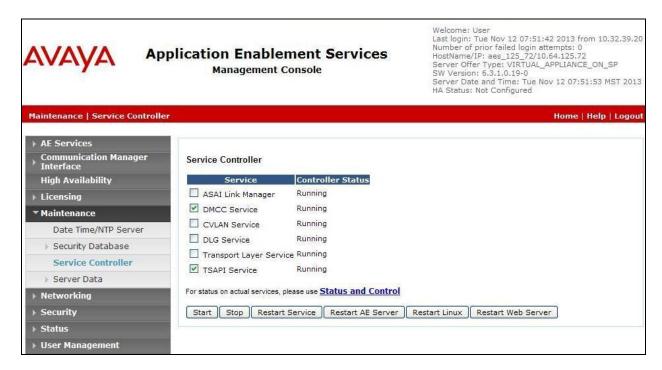
6.6. Administer TCP Settings

Select **Networking** → **TCP Settings** from the left pane, to display the **TCP Settings** screen in the right pane. For **TCP Retransmission Count**, select **TSAPI Routing Application Configuration**, as shown below.



6.7. Restart Services

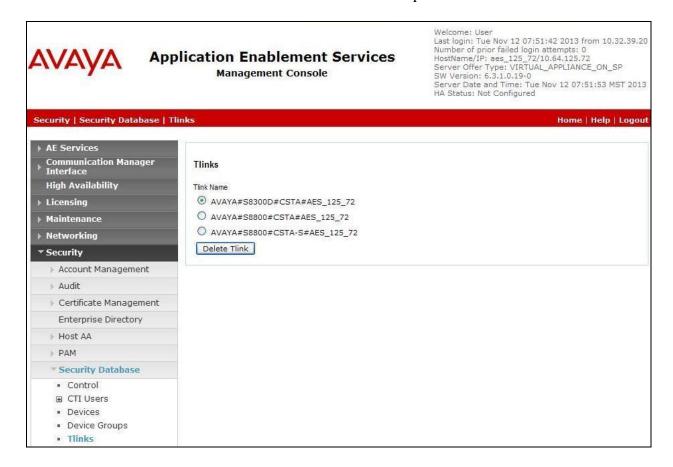
Select Maintenance \rightarrow Service Controller from the left pane, to display the Service Controller screen in the right pane. Check DMCC Service and TSAPI Service, and click Restart Service.



6.8. 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 Communications Center.

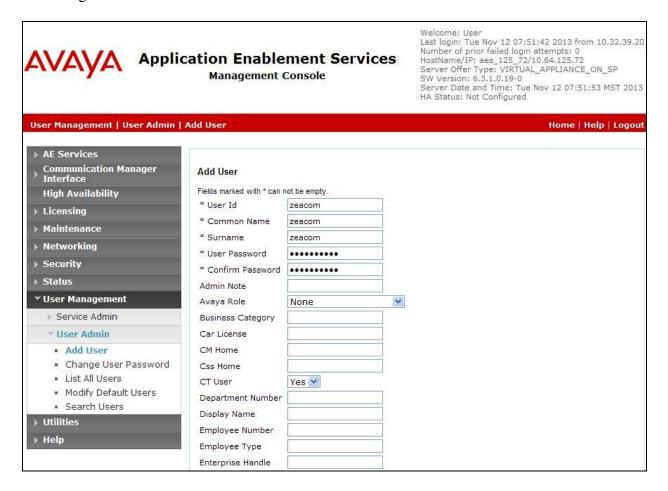
In this case, the associated Tlink name is "AVAYA#S8300D#CSTA#AES_125_72". Note the use of the switch connection "S8300D" from Section 6.3 as part of the Tlink name.



6.9. Administer Zeacom User

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

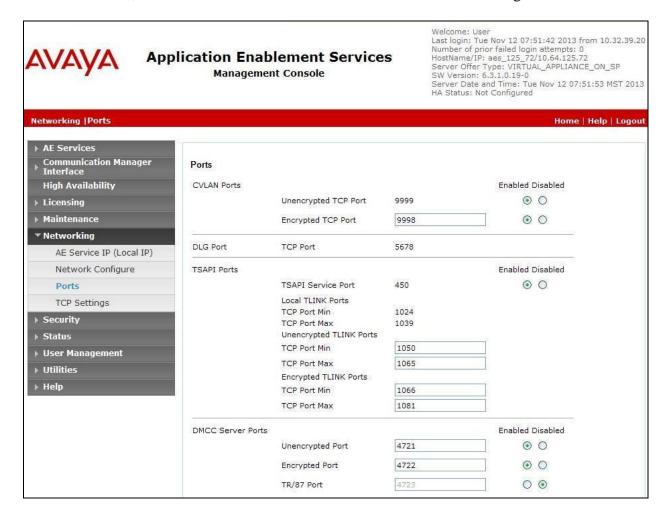
Enter desired values for **User Id**, **Common Name**, **Surname**, **User Password**, and **Confirm Password**. For **CT User**, select "Yes" from the drop-down list. Retain the default value in the remaining fields.



6.10. Enable 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.



7. Configure Zeacom Communications Center

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

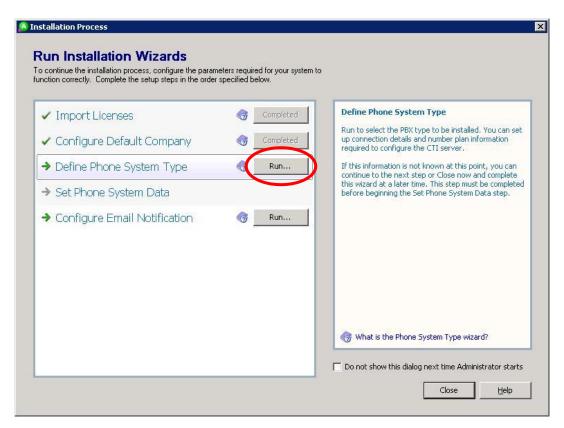
- Administer phone system type
- Administer phone system data
- Administer queues
- Administer agents and supervisors
- Administer mailboxes

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

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**.

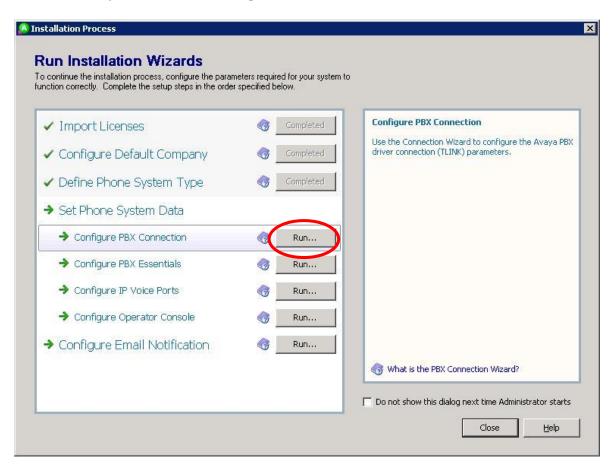


The **Phone System Type** screen is displayed. For **PBX Type**, select "Avaya Communication Manager (ACM)".

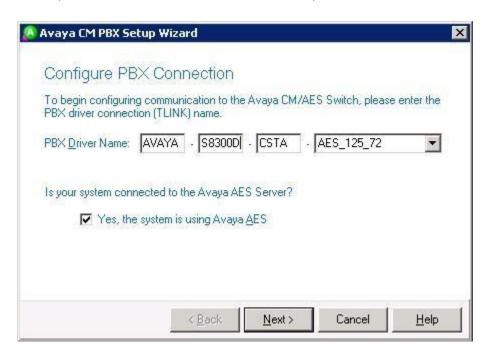


7.2. Administer Phone System Data

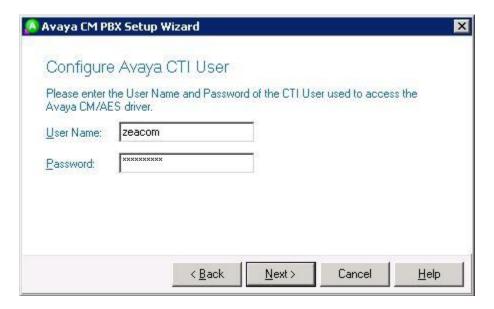
The **Installation Process** screen shown below is displayed next. Click the **Run** icon associated with **Set Phone System Data** → **Configure PBX Connection**.



The Avaya CM PBX Setup Wizard → Configure PBX Connection screen is displayed. For PBX Driver Name, enter the Tlink name from Section 6.3, as shown below.



The Avaya CM PBX Setup Wizard → Configure Avaya CTI User screen is displayed next. Enter the Zeacom user credentials from Section 6.9.



The Avaya CM PBX Setup Wizard \rightarrow 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.

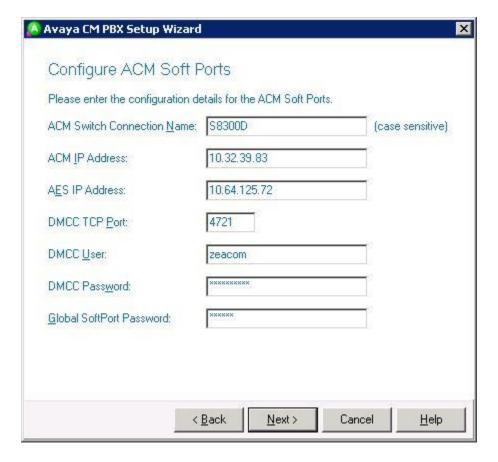
• **ACM IP Address:** IP address of H.323 gatekeeper from **Section 6.4**.

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

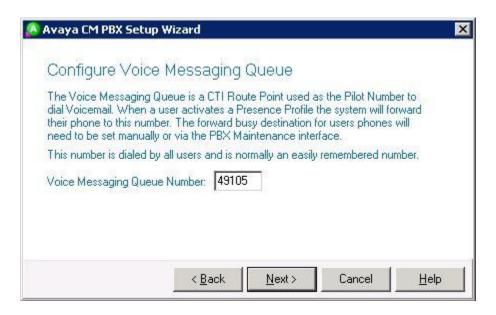
• **DMCC TCP Port:** "4721"

DMCC User: The Zeacom user credentials from Section 6.9.
 DMCC Password: The Zeacom user credentials from Section 6.9.

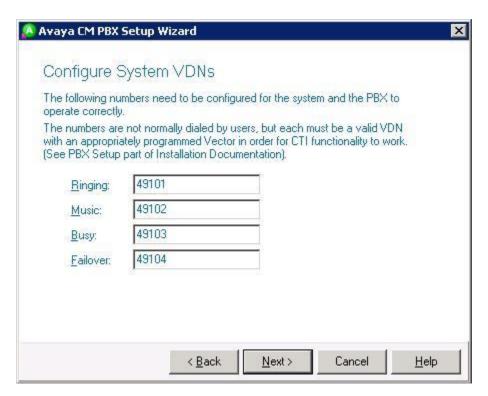
• Global SoftPort Password: The security code value from Section 5.6.



Continue with the Installation Wizard until the **Avaya CM PBX Setup Wizard** → **Configure Voice Messaging Queue** screen is displayed. For **Voice Messaging Queue Number**, enter the extension of the Voicemail VDN from **Section 5.3**.

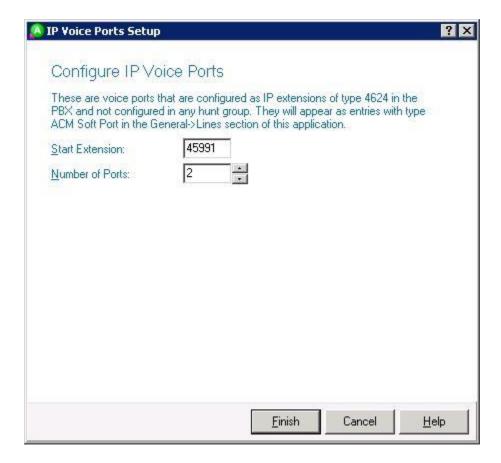


The Avaya CM PBX Setup Wizard → Configure System VDNs screen is displayed next. Enter the Ringing, Music, Busy, and Failure VDNs from Section 5.3, as shown below.



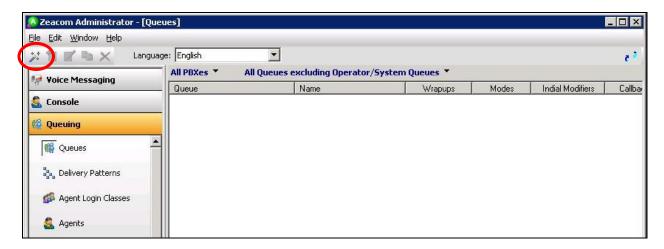
Continue with the Installation Wizard until the IP Voice Ports Setup → Configure IP Voice Ports screen is displayed. For Start Extension, enter the first virtual IP softphone extension from Section 5.6. For Number of Ports, enter the total number of virtual IP softphones from Section 5.6.

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.

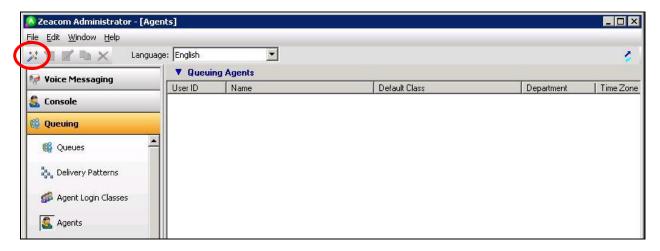


Follow the **Adding a New Queue Wizard** in the subsequent screens (not shown) to configure a new queue for each general routing VDN in **Section 5.3**. In the compliance testing, two queues were created as shown below.



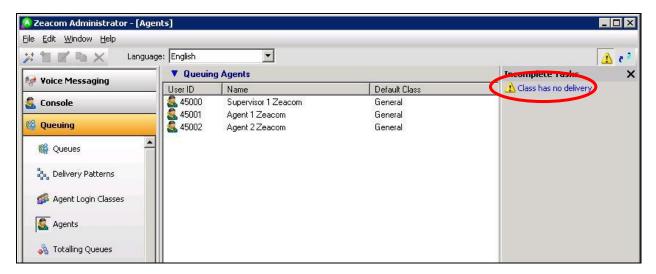
7.4. Administer Agents and Supervisors

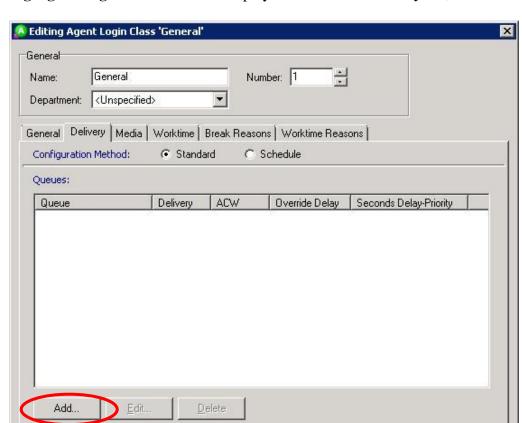
Select Queuing \rightarrow Agents from the left pane, followed by the Add Wizard icon located at the upper left corner of the screen.



Follow the **Add Agent Wizard** in the subsequent screens (not shown) to configure a corresponding entry for each agent and supervisor in **Section 3**. In the compliance testing, two agents and one supervisor were created as shown below.

Double click on Class has no delivery.





The **Editing Agent Login Class** screen is displayed. Select the **Delivery** tab, and click **Add**.

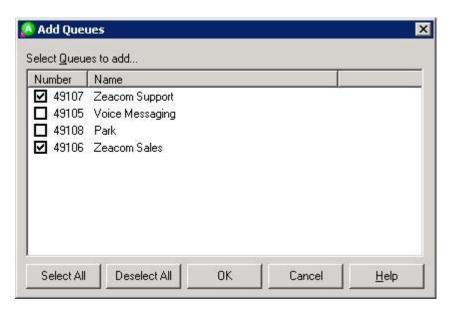
The **Add Queues** screen is displayed next. Check the entries corresponding to the general routing VDNs from **Section 5.3**, to enable calls to these VDNs to be delivered.

OK

Cancel

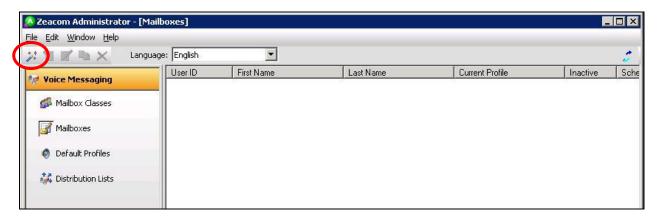
Apply

Help

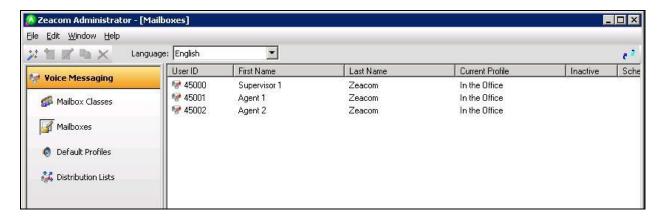


7.5. Administer Mailboxes

Select Voice Messaging \rightarrow 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 7.4**. The screen below shows the mailboxes that were created.



8. Verification Steps

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

8.1. Verify Avaya Aura® Communication Manager

On Communication Manager, verify the 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 6 no aes_125_72 established 98 67
```

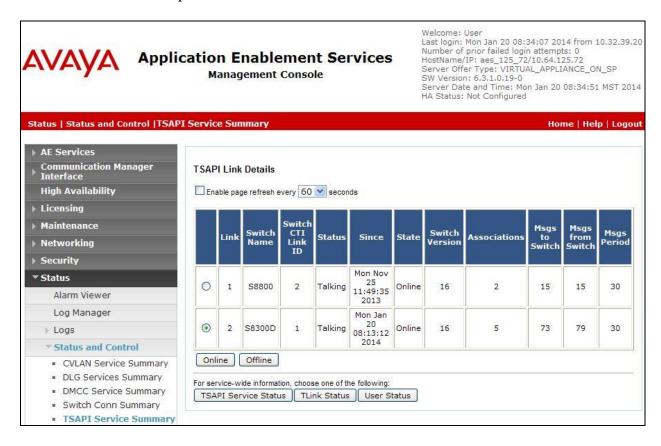
Verify the registration status of the virtual IP softphones by using the "list registered-ip-stations" command. Verify that all extensions from **Section 5.6** are displayed along with the IP address of the Application Enablement Services server, as shown below.

list registered-ip-stations				
		REGIST	ERED	IP STATIONS
Station Ext or Orig Port				Station IP Address/ Gatekeeper IP Address
45000	9650	IP_Phone	У	
	1	3.210A		10.32.39.83
45001	1608	IP_Phone	У	10.32.39.104
	1	1.340B		10.32.39.83
45002	9611	IP Phone	У	10.32.39.105
	1	6.3037		10.32.39.83
45991	4624	IP API A	У	10.64.125.72
	1	3.2040		10.32.39.83
45992	4624	IP API A	У	10.64.125.72
	1	3.2040		10.32.39.83

8.2. Verify Avaya Aura® Application Enablement Services

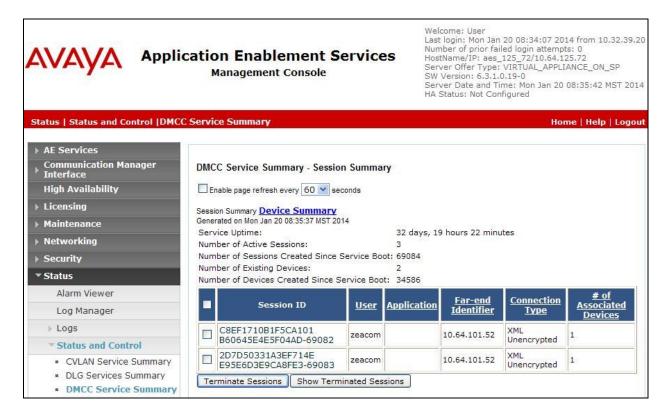
On Application Enablement Services, verify the status of the TSAPI link by selecting **Status Status and Control TSAPI Service Summary** from the left pane. The **TSAPI Link Details** screen is displayed.

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**.



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 Zeacom user name from **Section 6.9**, and that the total number of sessions reflects the number of virtual IP softphones from **Section 5.6**.



8.3. Verify Zeacom Communications Center

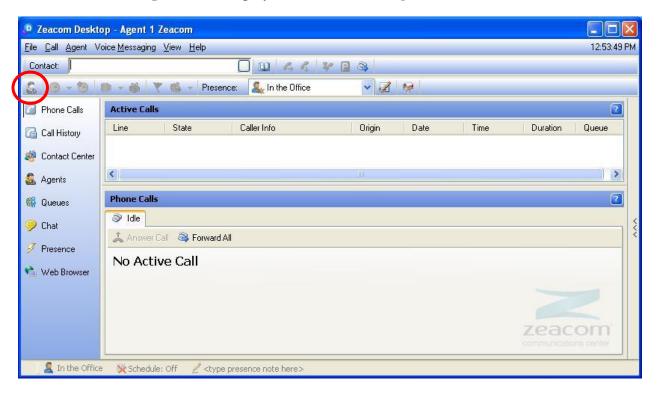
From the agent desktop running the Zeacom Executive Desktop client application, double-click on the **Desktop** icon shown below, which was created as part of installation.



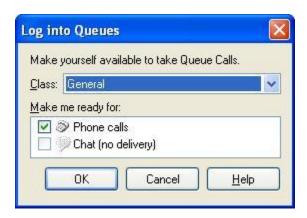
The **Desktop** login screen is displayed. Enter a valid login name from **Section 7.4**, and use the generic default PIN value from Zeacom. Retain the default value in the remaining fields.



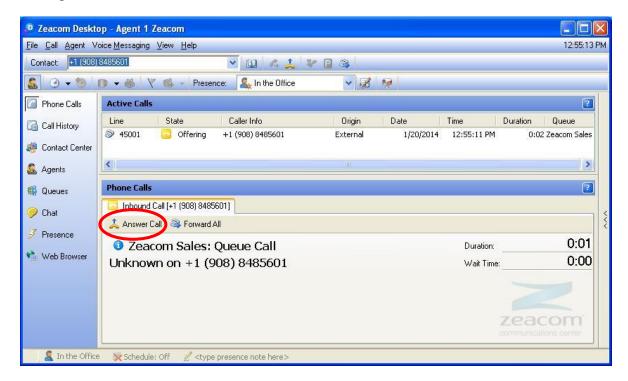
The **Zeacom Desktop** screen is displayed. Click on the **Log into Queues** icon shown below.



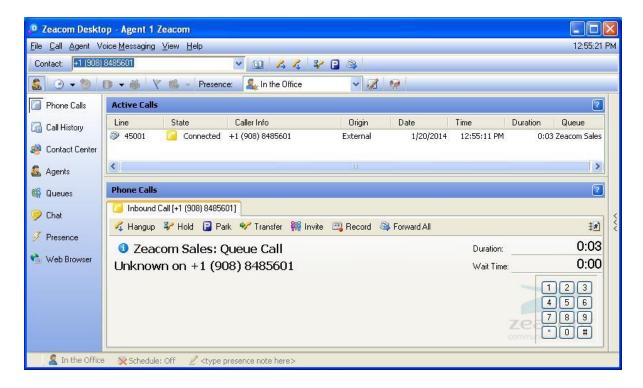
The **Log into Queues** dialog box is displayed next. Retain all default values.



Make an incoming call to the Zeacom Sales application, with available agent "45001". Verify that the agent desktop is populated with a voice call entry, as shown below, and that the **State** is "Offering". Click **Answer Call**.



Verify that the **State** is updated to "Connected", and that the agent is connected to the caller with two-way talk paths.



9. Conclusion

These Application Notes describe the configuration steps required for Zeacom Communications Center to successfully interoperate with Avaya Aura® Communication Manager using Avaya Aura® Application Enablement Services 6.3. All feature and serviceability test cases were completed with one observation 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*, Document 03-300509, Issue 9, Release 6.3, October 2013, available at http://support.avaya.com.
- **2.** Avaya Aura® Application Enablement Services Administration and Maintenance Guide, Release 6.3, Issue 2, October 2013, available at http://support.avaya.com.
- **3.** Communication Manager Installation Manual, Zeacom Library Version 7.0, available via Communication Manager training course provided by Zeacom.

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