

Avaya Solution & Interoperability Test Lab

Application Notes for TantaComm Capture with Avaya Aura® Communication Manager 6.3 and Avaya Aura® Application Enablement Services 6.3 using Single Step Conference – Issue 1.0

Abstract

These Application Notes describe the configuration steps required for TantaComm Capture to interoperate with Avaya Aura® Communication Manager and Avaya Aura® Application Enablement Services 6.3 using Single Step Conference. TantaComm Capture is a call recording solution.

In the compliance testing, TantaComm Capture used the Telephony Services Application Programming Interface and Device, Media, and Call Control XML interface from Avaya Aura® Application Enablement Services to monitor skill groups and agent stations on Avaya Aura® Communication Manager, and to capture the media associated with the monitored agents for call recordings.

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 TantaComm Capture to interoperate with Avaya Aura® Communication Manager and Avaya Aura® Application Enablement Services 6.3 using Single Step Conference. TantaComm Capture is a call recording solution.

In the compliance testing, TantaComm Capture used the Telephony Services Application Programming Interface (TSAPI) and Device, Media, and Call Control (DMCC) XML interface from Avaya Aura® Application Enablement Services to monitor skill groups and agent stations on Avaya Aura® Communication Manager, and to capture the media associated with the monitored agents for call recordings.

The TSAPI interface is used by TantaComm Capture to monitor skill groups and agent stations on Avaya Aura® Communication Manager. The DMCC interface is used by TantaComm Capture to register virtual IP softphones, and for adding virtual IP softphones to active calls using the Single Step Conference feature.

When there is an active call at the monitored agent, TantaComm Capture is informed of the call via event reports from the TSAPI interface. TantaComm Capture starts the call recording by using the Single Step Conference feature from the DMCC interface to add a virtual IP softphone to the active call to obtain the media. The event reports are also used to determine when to stop the call recordings.

2. General Test Approach and Test Results

The feature test cases were performed both automatically and manually. Upon start of the Capture application, the application automatically requests monitoring of skill groups and agent stations using TSAPI, and registers the virtual IP softphones using DMCC.

For the manual part of the testing, each call was handled manually on the agent station with generation of unique audio content for the recordings. Necessary user actions such as hold and resume were performed from the agent telephones to test the different call scenarios.

The serviceability test cases were performed manually by disconnecting/reconnecting the Ethernet connection to Capture.

The verification of tests included use of Capture logs for proper message exchanges, and use of Capture web interface for proper logging and playback of calls.

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 Capture:

- Handling of TSAPI messages in areas of event notification.
- Use of DMCC registration services to register and un-register the virtual IP softphones.
- Use of DMCC call control services to activate Single Step Conference for the virtual IP softphones to obtain media for call recordings.
- Proper recording, logging, and playback of calls for scenarios involving inbound, outbound, internal, external, ACD, non-ACD, hold, resume, multiple calls, multiple agents, conference, and transfer.

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

2.2. Test Results

All test cases were executed, and the following were observations on Capture:

- Capture version 14.0.0 displayed as 1.0.0 due to build problem. This will be addressed in build 14.0.1.
- For the attended transfer scenario involving agent transferring of call to non-monitored supervisor, the recording entry may not show up until the next call takes place in the system and with call duration reflecting the extra wait period. Nevertheless, the actual call is recorded properly up to the point of transfer as expected.
- For internal calls between two local users, by design the application produced one recording entry against the destination user when the destination user is monitored. As such, an internal call from a monitored agent to a non-monitored destination such as a supervisor was therefore not recorded. Similarly, for the attended conference scenario involving agent conferencing a non-monitored supervisor, the private conversation between the agent and the non-monitored supervisor was not recorded.

2.3. Support

Technical support on Capture can be obtained through the following:

Phone: (800) 444-8522, option 2Email: support@tantacomm.com

3. Reference Configuration

Capture can be configured on a single server or with components distributed across multiple servers. The compliance test used a single server configuration.

The detailed administration of basic connectivity between Communication Manager, Application Enablement Services, System Manager, Session Manager, and of contact center devices are not the focus of these Application Notes and will not be described.

In the compliance testing, Capture monitored the skill groups and agent stations shown in the table below.

Device Type	Extension
VDN	49001, 49002
Skill Group	48101, 48102
Supervisor	45000
Agent Station	45001, 46002
Agent ID	45881, 45882

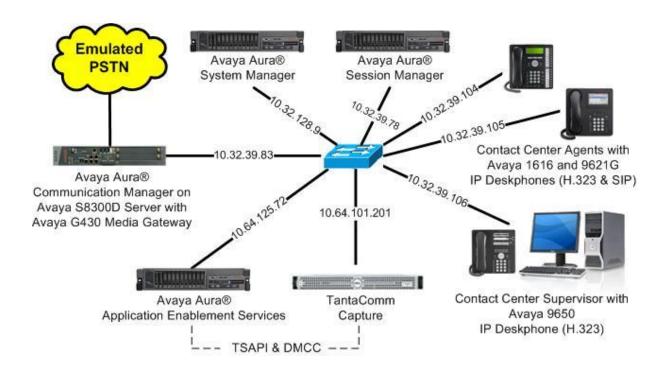


Figure 1: Compliance Testing Configuration

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	6.3.5 (R016x.03.0.124.0-21460)
Avaya G430 Media Gateway	6.3.5 (35.8.0)
Avaya Aura® Application Enablement Services	6.3.1 (6.3.1.0.19-0)
Avaya Aura® Session Manager	6.3.7
Avaya Aura® System Manager	6.3.5
Avaya 1616 IP Deskphone (H.323)	1.350B
Avaya 9621G IP Deskphone (SIP)	6.3.1.22
Avaya 9650 IP Deskphone (H.323)	3.220A
TantaComm Capture on	14.0.0
Windows Server 2008	R2 Standard
• TStsapi.exe	14.2.0.0
ars_dmcc.exe	14.2.2.0
 Avaya TSAPI Windows Client (csta32.dll) 	6.1.1.469
Avaya DMCC XML SDK	6.1.0.501

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 system parameters features
- Administer virtual IP softphones
- Administer IP codec set

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.

```
display system-parameters customer-options
                                                                                Page 3 of 11
                                        OPTIONAL FEATURES
    Abbreviated Dialing Enhanced List? y
Access Security Gateway (ASG)? n
Analog Trunk Incoming Call ID? y
CAS Branch? n
Cap/Sys List Dialing Start at 01? y
Wer Supervision by Call Classifier? y
Audible Message Waiting? y
Authorization Codes? y
CAS Branch? n
CAS Main? n
Change COR by FAC? n
A/D Grp/Sys List Dialing Start at 01? y
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
                                                      DCS Call Coverage? y
            ARS/AAR Dialing without FAC? y
            ASAI Link Core Capabilities? n
           ASAI Link Plus Capabilities? n
                                                                    DCS with Rerouting? y
        Async. Transfer Mode (ATM) PNC? n
  Async. Transfer Mode (ATM) Trunking? n Digital Loss Plan Modification? y
                ATM WAN Spare Processor? n
                                                                 DS1 Echo Cancellation? y
                                         ATMS? y
                      Attendant Vectoring? 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

CTI Link: 1

Extension: 40001

Type: ADJ-IP

COR: 1

Name: AES CTI Link
```

5.3. Administer System Parameters Features

Use the "change system-parameters features" command to enable **Create Universal Call ID** (**UCID**), which is located on **Page 5**. For **UCID Network Node ID**, enter an available node ID.

```
change system-parameters features
                                                               Page
                                                                      5 of 20
                       FEATURE-RELATED SYSTEM PARAMETERS
SYSTEM PRINTER PARAMETERS
 Endpoint:
                        Lines Per Page: 60
SYSTEM-WIDE PARAMETERS
                                    Switch Name:
           Emergency Extension Forwarding (min): 10
         Enable Inter-Gateway Alternate Routing? n
Enable Dial Plan Transparency in Survivable Mode? n
                             COR to Use for DPT: station
               EC500 Routing in Survivable Mode: dpt-then-ec500
MALICIOUS CALL TRACE PARAMETERS
              Apply MCT Warning Tone? n MCT Voice Recorder Trunk Group:
     Delay Sending RELease (seconds): 0
SEND ALL CALLS OPTIONS
    Send All Calls Applies to: station
                                          Auto Inspect on Send All Calls? n
             Preserve previous AUX Work button states after deactivation? n
UNIVERSAL CALL ID
    Create Universal Call ID (UCID)? y
                                          UCID Network Node ID: 1
```

Navigate to **Page 13**, and enable **Send UCID to ASAI**. This parameter allows for the universal call ID to be sent to Capture.

```
change system-parameters features
                                                               Page 13 of 20
                       FEATURE-RELATED SYSTEM PARAMETERS
CALL CENTER MISCELLANEOUS
          Callr-info Display Timer (sec): 10
                         Clear Callr-info: next-call
       Allow Ringer-off with Auto-Answer? n
   Reporting for PC Non-Predictive Calls? n
           Agent/Caller Disconnect Tones? n
         Interruptible Aux Notification Timer (sec): 3
            Zip Tone Burst for Callmaster Endpoints: double
 ASAI
            Copy ASAI UUI During Conference/Transfer? y
       Call Classification After Answer Supervision? y
                                   Send UCID to ASAI? y
         For ASAI Send DTMF Tone to Call Originator? y
 Send Connect Event to ASAI For Announcement Answer? n
```

5.4. Administer Virtual IP Softphones

Add a virtual IP 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.

• **Extension:** The available extension number.

• **Type:** "4624"

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

• IP SoftPhone: "y"

```
add station 45991
                                                            Page
                                                                  1 of
                                  STATION
Extension: 45991
                                      Lock Messages? n
                                                                  BCC: 0
    Type: 4624
                                      Security Code: 123456
                                                                   TN: 1
                                                                 COR: 1
    Port: IP
                                    Coverage Path 1:
    Name: TantaComm Virtual #1
                                    Coverage Path 2:
                                                                  cos: 1
                                   Hunt-to Station:
                                                                Tests: y
STATION OPTIONS
                                        Time of Day Lock Table:
            Loss Group: 19 Personalized Ringing Pattern: 1
      Speakerphone: 2-way

Display Language: english

able GK Node Name:
                                             Message Lamp Ext: 45991
Survivable GK Node Name:
      Survivable COR: internal
                                             Media Complex Ext:
                                                  IP SoftPhone? v
  Survivable Trunk Dest? v
                                            IP Video Softphone? n
                            Short/Prefixed Registration Allowed: default
                                           Customizable Labels? Y
```

Repeat this section to administer the desired number of virtual IP softphones. In the compliance testing, two virtual IP softphones were administered as shown below, to allow for simultaneous recording of two monitored agents in **Section 3**.

```
list station 45991 count 2
                            STATIONS
                                         Room/ Cv1/ COR/
Ext/
          Port/ Name/
           Type Surv GK NN Move Data Ext Cv2 COS TN Jack
Hunt-to
45991
           S00051 TantaComm Virtual #1
                                                        1
            4624
                                                        1
                                                            1
                                    no
           S00054 TantaComm Virtual #2
45992
                                                        1
            4624
```

5.5. Administer IP Codec Set

Use the "change ip-codec-set n" command, where "n" is an existing codec set number used for integration with Capture. For **Audio Codec**, make sure "G.729A" is included, as this is the only codec type supported by Capture. In the compliance testing, this IP codec set was assigned to the agents and to the virtual IP softphones.

```
change ip-codec-set 1
                                                                1 of
                                                                      2
                                                          Page
                       IP Codec Set
   Codec Set: 1
   Audio
              Silence
                          Frames
                                   Packet
   Codec
               Suppression Per Pkt Size(ms)
1: G.711MU
                                    20
                           2
                   n
2: G.729A
                   n
                            2
                                     20
 3:
```

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
- Restart services
- Obtain Tlink name
- Administer TantaComm 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.



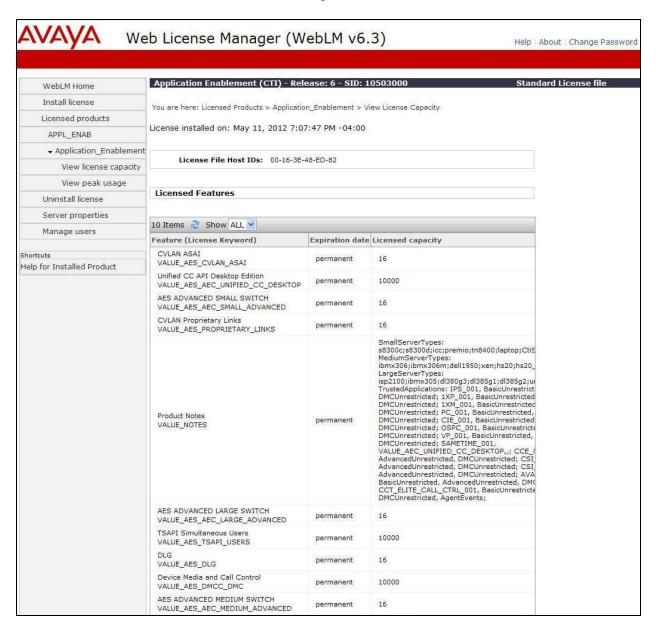
6.2. Verify License

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.



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**.

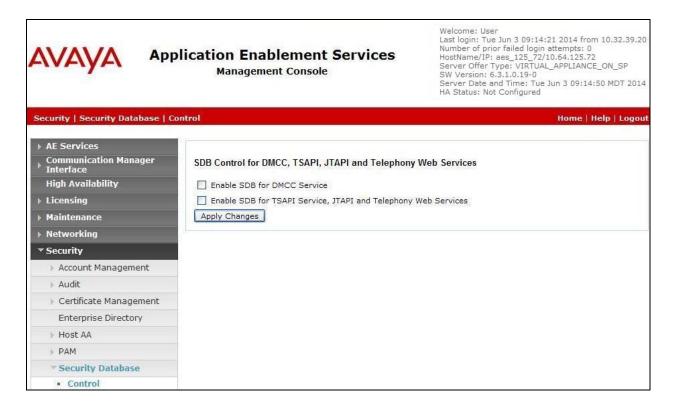


The **Edit H.323 Gatekeeper** screen is displayed. Enter the IP address of a C-LAN circuit pack or the Processor C-LAN on Communication Manager to use as the H.323 gatekeeper, in this case "10.32.39.83" as shown below. Click **Add Name or IP**.



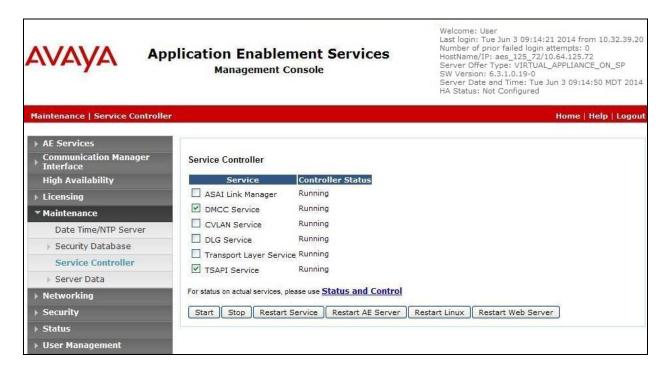
6.5. Disable Security Database

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. Restart Services

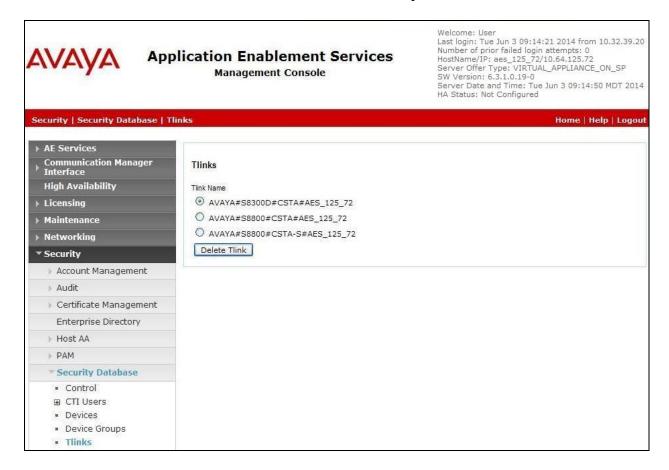
Select Maintenance \rightarrow Service Controller from the left pane, to display the Service Controller screen in the right pane. Check DMCC Service and TSAPI Service, and click Restart Service.



6.7. 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 Capture.

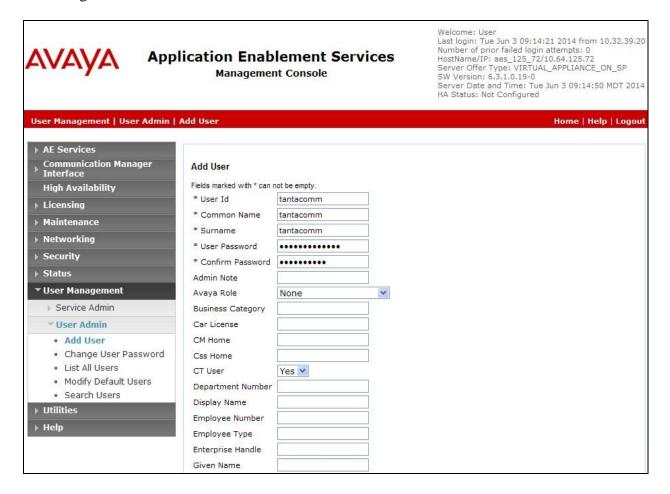
In this case, the associated Tlink name is "AVAYA#S8300D#CSTA#AES_125_72". Note the use of the switch connection "S8300D" from **Section 6.3** as part of the Tlink name.



6.8. Administer TantaComm User

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

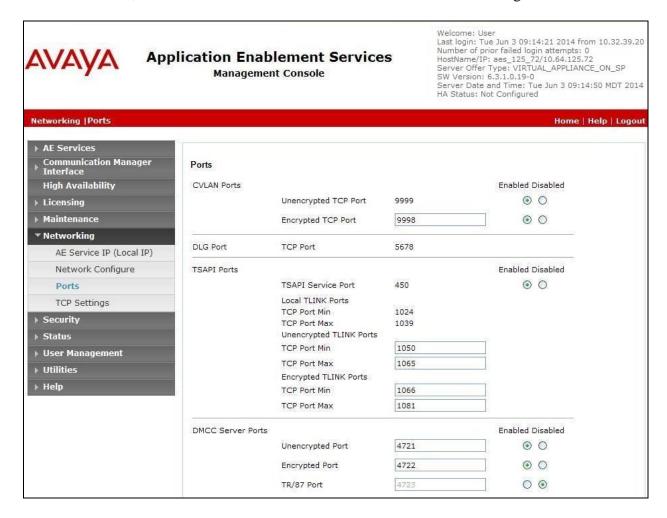
Enter desired values for **User Id**, **Common Name**, **Surname**, **User Password**, and **Confirm Password**. For **CT User**, select "Yes" from the drop-down list. Retain the default value in the remaining fields.



6.9. 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 Avaya Aura® Session Manager

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

- Launch System Manager
- Administer users

7.1. Launch System Manager

Access the System Manager web interface by using the URL "https://ip-address" in an Internet browser window, where "ip-address" is the IP address of System Manager. Log in using the appropriate credentials.



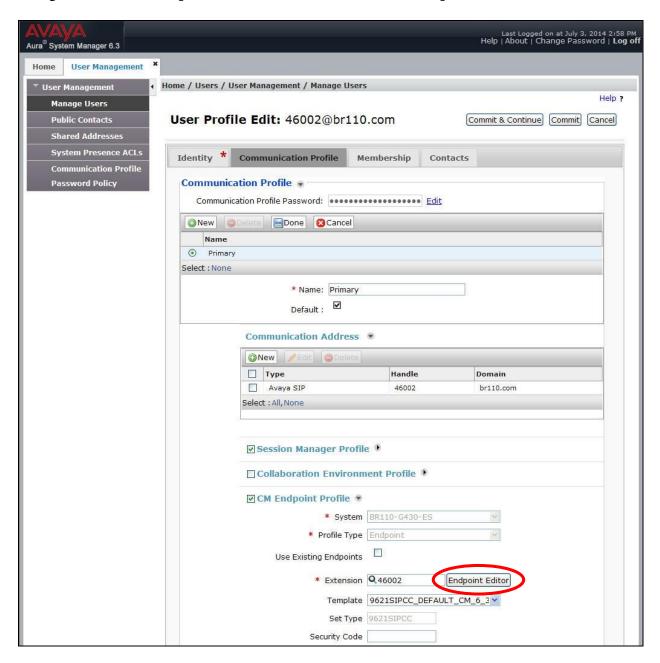
7.2. Administer Users

In the subsequent screen (not shown), select Users → User Management → Manage Users to display the User Management screen below. Select the entry associated with the first SIP agent station from Section 3, in this case "46002", and click Edit.



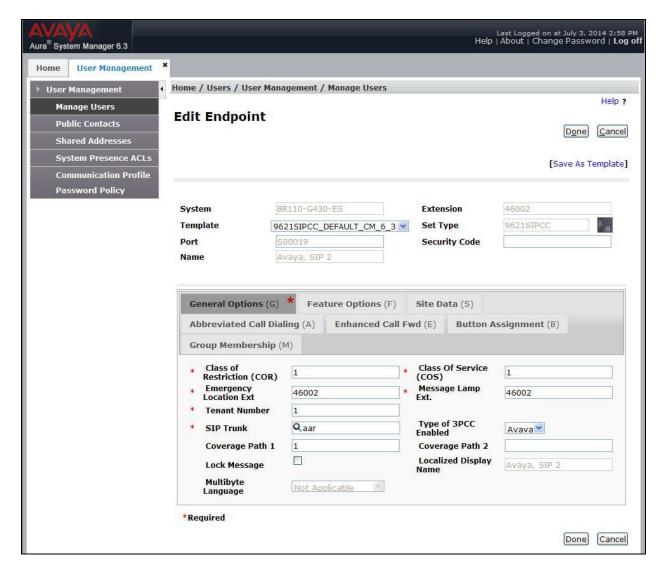
The **User Profile Edit** screen is displayed. Select the **Communication Profile** tab to display the screen below.

Navigate to the CM Endpoint Profile sub-section, and click Endpoint Editor.



The **Edit Endpoint** screen is displayed next. For **Type of 3PCC Enabled**, select "Avaya" from the drop-down list as shown below. Retain the default values in the remaining fields.

Repeat this section for all SIP agent users.



8. Configure TantaComm Capture

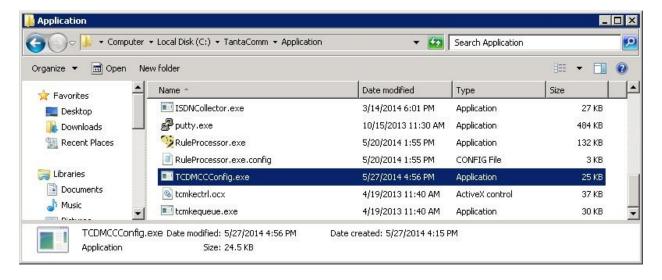
This section provides the procedures for configuring Capture. The procedures include the following areas:

- Administer TCDMCCConfig
- Restart services

The configuration of Capture is performed by the TantaComm technical services team. The procedural steps are presented in these Application Notes for informational purposes.

8.1. Administer TCDMCCConfig

From the Capture server, navigate to the **C:\TantaComm\Application** directory, and double click on the **TCDMCCConfig** application shown below.



The **TantaComm Configuration for Avaya DMCC** screen is displayed. Enter the following values for the specified fields.

TantaSwitch: Select the applicable switch that was pre-configured.
 AES IP: Enter the IP address of Application Enablement Services.

• **Connection String:** The Tlink name from **Section 6.7**.

• **TSAPI Port:** "450"

TSAPI User: The TantaComm user credentials from Section 6.8.
 TSAPI Password: The TantaComm user credentials from Section 6.8.

• DMCC Recording Server: "Local"

• **Server Name:** Enter a desired name.

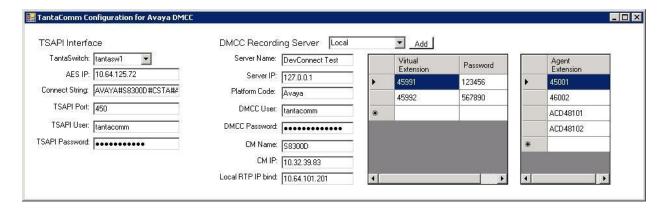
Server IP: "127.0.0.1"Platform Code: "Avaya"

DMCC User: The TantaComm user credentials from Section 6.8.
 DMCC Password: The TantaComm user credentials from Section 6.8.
 CM Name: The switch connection name from Section 6.3.
 CM IP: The TantaComm user credentials from Section 6.8.
 The switch connection name from Section 6.4.

• **Local RTP IP bind:** The IP address of the Capture server.

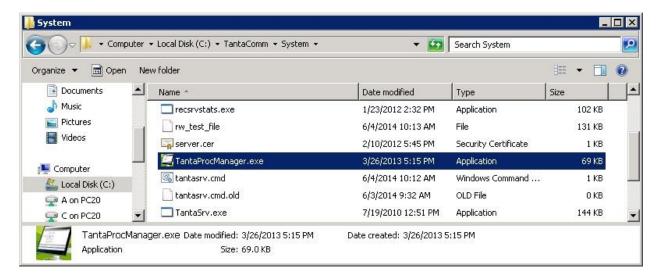
In the **Virtual Extension** and **Password** columns, enter the extension and corresponding security code of each virtual IP softphone from **Section 5.4**.

In the **Agent Extension** column, enter the agent station and skill group extensions from **Section** 3. For skill group extensions, prepend "ACD" before the extension as shown below.

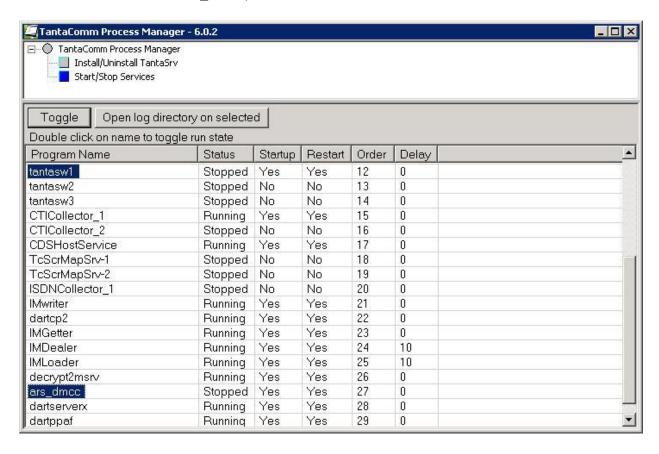


8.2. Restart Services

Navigate to C:\TantaComm\System directory, and double click on the TantaProcManager application shown below.



The **TantaComm Process Manager** screen is displayed. Scroll down as necessary to select and restart the **tantasw1** and **ars dmcc**, as shown below.



9. Verification Steps

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

9.1. 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 57 57
```

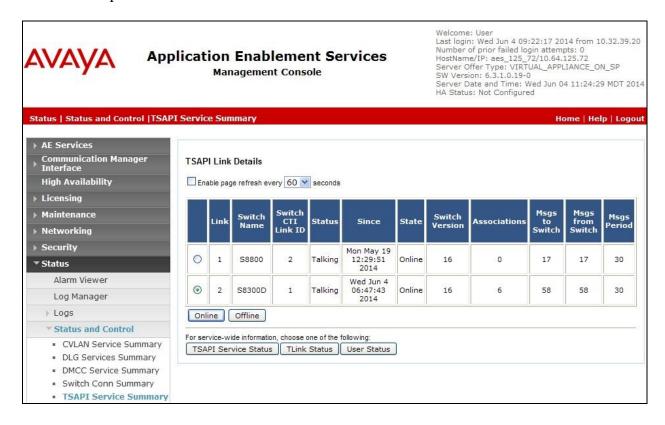
Verify the registration status of the virtual IP softphones by using the "list registered-ip-stations" command. Verify that all virtual IP softphone extensions from **Section 5.4** are displayed along with the IP address of the Application Enablement Services server, as shown below.

list registered-ip-stations					
REGISTERED IP STATIONS					
Station Ext or Orig Port				Station IP Address/ Gatekeeper IP Address	
45000	9650 1	IP_Phone 3.220A	У	10.32.39.106 10.32.39.83	
45001	1616 1	IP_Phone 1.350B	У	10.32.39.104 10.32.39.83	
45991	4624 1	IP_API_A 3.2040	У	10.64.125.72 10.32.39.83	
45992	4624 1	IP_API_A 3.2040	У	10.64.125.72 10.32.39.83	

9.2. Verify Avaya Aura® Application Enablement Services

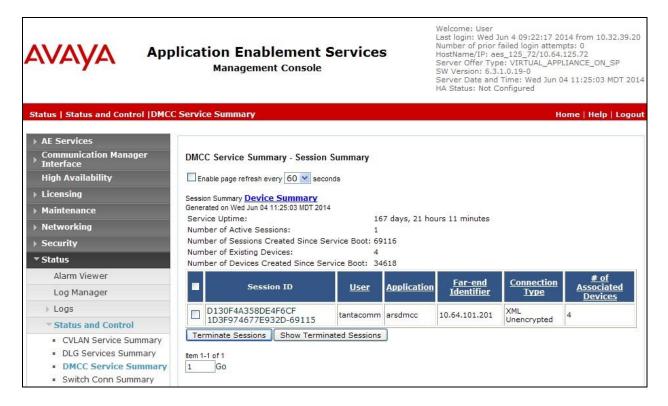
On Application Enablement Services, verify the status of the TSAPI link by selecting **Status Status and Control TSAPI Service Summary** from the left pane. The **TSAPI Link Details** screen is displayed.

Verify the **Status** is "Talking" for the TSAPI link administered in **Section 6.3**, and that the **Associations** column reflects the total number of monitored skill groups, agent stations, and virtual IP softphones from **Section 8.1**.



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 an active session with the TantaComm user name from **Section 6.8**, and that the # **of Associated Devices** column reflects the total number of virtual IP softphone and agent station extensions from **Section 8.1**.

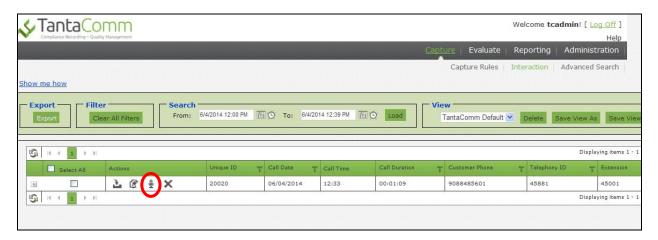


9.3. Verify TantaComm Capture

Log an agent into the skill group to handle and complete an ACD call. Access the Capture web-based interface by using the URL "http://ip-address/capture" in an Internet browser window, where "ip-address" is the IP address of the Capture server. Log in using the appropriate credentials.

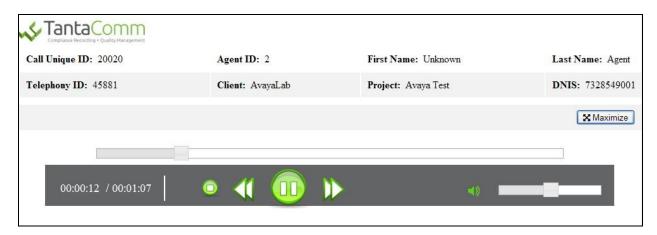


The **TantaComm** screen below is displayed. Set the applicable **Search** date and time range to display a list of recent recording entries. Verify that there is an entry reflecting the last call, with proper values in the relevant fields. Click on the associated **Play Audio** icon shown below.



Verify that a pop-up screen is displayed and that the call recording is played back.

Note that the **Project** and **Client** names shown below were all pre-configured parameters, and that the agent **First Name** and **Last Name** can be configured on the Capture server if desired for display purposes. The agent names shown below are the default values.



10. Conclusion

These Application Notes describe the configuration steps required for TantaComm Capture to successfully interoperate with Avaya Aura® Communication Manager and Avaya Aura® Application Enablement Services 6.3 using Single Step Conference. All feature and serviceability test cases were completed with an observation noted in **Section 2.2**.

11. 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.** *CAPTURE VOIP Recording with DMCC*, 2013, available upon request to TantaComm support.
- **4.** *TantaComm Capture Administration guides*, available upon request to TantaComm support.

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