

## **DevConnect Program**

Application Notes for Globitel SpeechLog Call Recording 8.0 with Avaya Aura® Communication Manager and Avaya Aura® Application Enablement Services 10.1 using DMCC Multiple Registration – Issue 1.0

## **Abstract**

These Application Notes describe the configuration steps required for Globitel SpeechLog Call Recording 8.0 to interoperate with Avaya Aura® Communication Manager 10.1 and Avaya Aura® Application Enablement Services 10.1. Globitel SpeechLog Call Recording is a voice recording solution which can be used to record voice streams for Avaya telephony using Multiple Registration recording method.

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.

## 1. Introduction

These Application Notes describe the configuration steps required for Globitel SpeechLog Call Recording 8.0 to interoperate with Avaya Aura® Communication Manager 10.1 and Avaya Aura® Application Enablement Services 10.1 using the Multiple Device Registration recording method.

Globitel SpeechLog Call Recording 8.0 is a voice recording system which can be used to record the voice stream of Avaya telephony endpoints. In this compliance test, it uses Avaya Aura® Communication Manager's Multiple Device Registration feature via Avaya Aura® Application Enablement Services (AES) Device, Media, and Call Control (DMCC) interface to capture the audio and call details for call recording. The application uses the Avaya Aura® Application Enablement Services DMCC service to register the extensions that are to be recorded. When the extension receives an event pertaining to the start of a call, the application receives the extensions RTP media stream.

# 2. General Test Approach and Test Results

The general test approach was to validate correct recording of calls in a variety of call handling scenarios and recovery from network interruption using DMCC Multiple Registration. All test cases were performed manually.

For the manual part of the testing, each call was handled manually on the extension telephone with generation of unique audio content for the recordings. Necessary user actions such as hold and reconnect 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 Globitel SpeechLog Call Recording.

DevConnect Compliance Testing is conducted jointly by Avaya and DevConnect members. The jointly defined test plan focuses on exercising APIs and/or standards-based interfaces pertinent to the interoperability of the tested products and their functionalities. DevConnect Compliance Testing is not intended to substitute full product performance or feature testing performed by DevConnect members, nor is it to be construed as an endorsement by Avaya of the suitability or completeness of a DevConnect member's solution.

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

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

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

## 2.1. Interoperability Compliance Testing

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

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

#### 2.2. Test Results

All test cases were executed and verified successfully.

## 2.3. Support

Technical support on Globitel SpeechLog Call Recording 8.0 can be obtained through the following: Globitel, Khalda, Amman, Jordan.

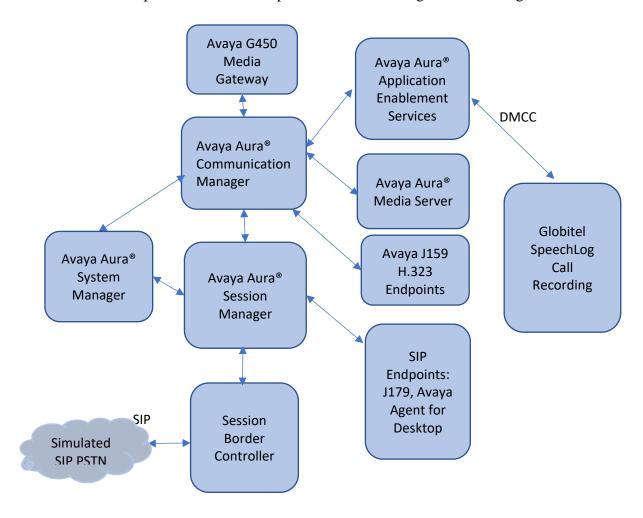
support@globitel.com

Hotline: +962 (7) 97315050 Phone: +962 (6) 5300 130 Fax: +962 (6) 5300 144

P.O. Box 1786 Amman 11821 Jordan

# 3. Reference Configuration

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



**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® System Manager in Virtual Environment	10.1.2.1012015476
Avaya Aura® Session Manager in Virtual Environment	10.1.2.0.1012016
Avaya Aura® Communication Manager in Virtual Environment	10.1.2 - 01.0.974.0 27783
Avaya G450 Media Gateway	42.18.1
Avaya Aura® Media Server in Virtual Environment	10.1.0 SP 2
Avaya Aura® Application Enablement Services in Virtual Environment	10.1.2.0.0.12-0
Avaya Session Border Controller in Virtual Environment	10.1.0.0-32-21432
Avaya Agent for Desktop for Windows	2.0.6.26.3001
Avaya J179 IP Phone (SIP)	4.1.1
Avaya J159 IP Deskphone (H.323)	6.8.5
Globitel SpeechLog Call Recording	8.0

# 5. Configure Avaya Aura® Communication Manager

The detailed administration of basic connectivity between Communication Manager and Application Enablement Services, and of contact center devices are not the focus of these Application Notes and will not be described. This section provides the procedures for configuring Communication Manager. The procedures include the following areas:

- Administer CTI link.
- Configure H.323 Stations for Multi-Registration.
- Configure SIP Stations for Multiple Registration.

#### 5.1. 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

CTI Link: 1

Extension: 79999

Type: ADJ-IP

COR: 1

Name: aes140

Unicode Name? n

## 5.2. Configure H.323 Stations for Multi-Registration

All endpoints that are to be monitored by Globitel will need to have IP Softphone set to **y**. IP Softphone must be enabled in order for Multiple Registration to work. Type **change station x** where **x** is the extension number of the station to be monitored. Also, note this extension number for configuration required during the Globitel setup in **Section 7**. Note the **Security Code** and ensure that **IP Softphone** is set to **y**.

```
change station 70010
                                                                    Page 1 of
                                                                                   5
                                       STATION
                                         Security Code: 111222
Coverage Path 1:
Coverage Path 2:
Hunt-to Station:
                                                                            BCC: 0
Extension: 70010
     Type: 9641
                                                                             TN: 1
     Port: S000004
                                                                            COR: 1
     Name: H323 Ext1
                                                                             cos: 1
                                                                          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: 70010
                                               Mute Button Enabled? y
                                                  Button Modules: 0
Survivable GK Node Name:
          Survivable COR: internal
                                                 Media Complex Ext:
  Survivable Trunk Dest? y
                                                        IP SoftPhone? y
                                                  IP Video Softphone? n
                                Short/Prefixed Registration Allowed: default
                                                 Customizable Labels? Y
```

For compliance testing, two H323 extensions were administered: **70010** and **70011**.

## 5.3. Configure SIP Stations for Multiple Registration

The configuration for SIP Stations should be using System Manager and this section only shows what needs to be changed.

Each Avaya SIP endpoint or station that needs to be monitored for call recording will need to have **Type of 3PCC Enabled** is set to **Avaya** and **IP Softphone** set to **Yes**.

display station 70000 Page 1 of STATION Security Code: 111222
Coverage Path 1:
Coverage Path 2:
Hunt-to Station: BCC: 0 Extension: 70000 Lock Messages? n Type: J179 TN: 1 Port: S000010 COR: 1 Name: SIP Ext1 cos: 1 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: 70000 Mute Button Enabled? y Button Modules: 0 Survivable GK Node Name: Survivable COR: internal Media Complex Ext: Survivable Trunk Dest? y IP SoftPhone? y IP Video Softphone? n Short/Prefixed Registration Allowed: default Customizable Labels? Y

#### Go to Page 6.

```
display station 70000
                                                                 Page 6 of
                                     STATION
SIP FEATURE OPTIONS
    Type of 3PCC Enabled: Avaya SIP Trunk: aar
Enable Reachability for Station Domain Control: s
SIP URI: 70000@aura.com
Primary Session Manager
IPv4 Address: 10.128.224.18 IPv6 Address: IPv4 Node Name: smsip18 IPv6 Node Name:
Secondary Session Manager
 IPv4 Address:
                                  IPv6 Address:
IPv4 Node Name:
                                  IPv6 Node Name:
Third Session Manager
 IPv4 Address:
                                 IPv6 Address:
IPv4 Node Name:
                                  IPv6 Node Name:
Fourth Session Manager
 IPv4 Address:
                                 IPv6 Address:
IPv4 Node Name:
                                  IPv6 Node Name:
```

For compliance testing, two SIP extensions were administered: 70000 and 70001.

# 6. Configure Avaya Aura® Application Enablement Services

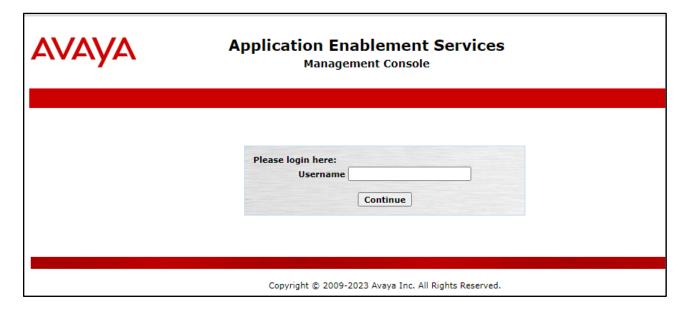
This section provides the procedures for configuring Application Enablement Services. The procedures include the following areas:

- Launch OAM interface
- Verify license
- Administer TSAPI link
- Administer Globitel user
- Enable CTI User
- Display Port Information
- Restart services

#### 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.



# AVAYA Application Enablement Services

**Management Console** 

Welcome: User cust Last login: Fri Apr 7 11:07:27 I.T. 2023 from 172.16.8.167 Number of prior failed login attempts: 0 HostName/IP: aes140.aura.com/10.30.5.140 Server Offer Type: VIRTUAL\_APPLIANCE\_ON\_VMWARE SW Version: 10.1.2.0.0.12-0 Server Date and Time: Fri Apr 07 11:08:30 ICT 2023

Home | Help | Logout

HA Status: Not Configured

Home **▶ AE Services** Communication Manager Interface **High Availability** ▶ Licensing ▶ Maintenance ▶ Networking **▶ Security** ▶ Status **▶ User Management** Utilities ▶ Help

#### Welcome to OAM

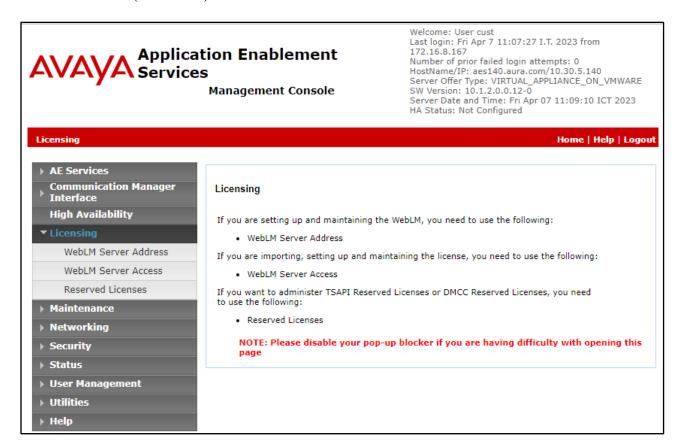
The AE Services Operations, Administration, and Management (OAM) Web provides you with tools for managing the AE Server. OAM spans the following administrative domains:

- . AE Services Use AE Services to manage all AE Services that you are licensed to use on the AE Server.
- · Communication Manager Interface Use Communication Manager Interface to manage switch connection and dialplan.
- High Availability Use High Availability to manage AE Services HA.
- Licensing Use Licensing to manage the license server.
   Maintenance Use Maintenance to manage the routine maintenance tasks.
- Networking Use Networking to manage the network interfaces and ports.
- Security Use Security to manage Linux user accounts, certificate, host authentication and authorization, configure Linux-PAM (Pluggable Authentication Modules for Linux) and so on.
- Status Use Status to obtain server status informations.
- User Management Use User Management to manage AE Services users and AE Services user-related resources.
- Utilities Use Utilities to carry out basic connectivity tests.
- . Help Use Help to obtain a few tips for using the OAM Help system

Depending on your business requirements, these administrative domains can be served by one administrator for all domains, or a separate administrator for each domain.

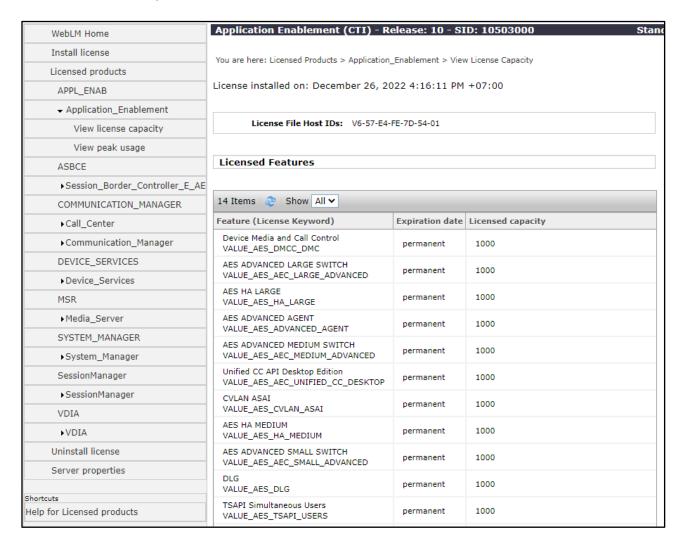
## 6.2. Verify License

Select **Licensing WebLM Server Access** in the left pane, to display the applicable WebLM server log in screen (not shown). Log in using the appropriate credentials and navigate to display installed licenses (not shown).



Select Licensed products  $\rightarrow$  APPL\_ENAB  $\rightarrow$  Application\_Enablement in the left pane, to display the Licensed Features screen in the right pane.

Verify that there are sufficient licenses for **Device Media and Call Control** and **TSAPI Simultaneous Users**, as shown below.

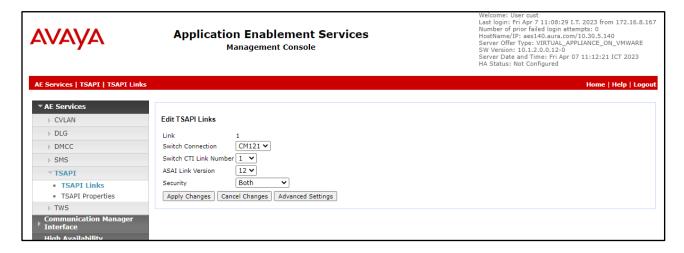


#### 6.3. Administer TSAPI Link

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**.



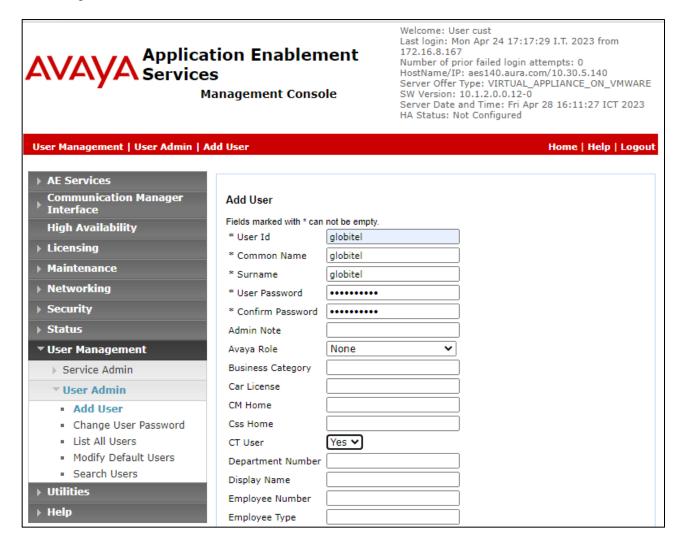
The **Add TSAPI Links** screen is displayed next. The **Link** field is only local to the Application Enablement Services server and may be set to any available number. For **Switch Connection**, select the relevant switch connection from the drop-down list. In this case, the existing switch connection **CM121** is selected. For **Switch CTI Link Number**, select the CTI link number from **Section 5.1**. Retain the default values in the remaining fields.



#### 6.4. Administer Globitel 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.

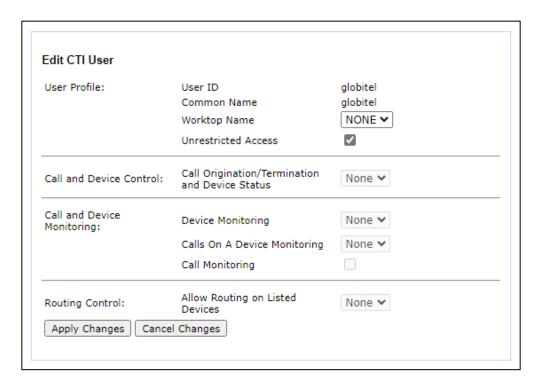


#### 6.5. Enable CTI User

Navigate to the CTI Users screen by selecting **Security**  $\rightarrow$  **Security Database**  $\rightarrow$  **CTI Users**  $\rightarrow$  **List All Users**. In the CTI Users window, select the user that was set up in **Section 6.4** and select the **Edit** option.

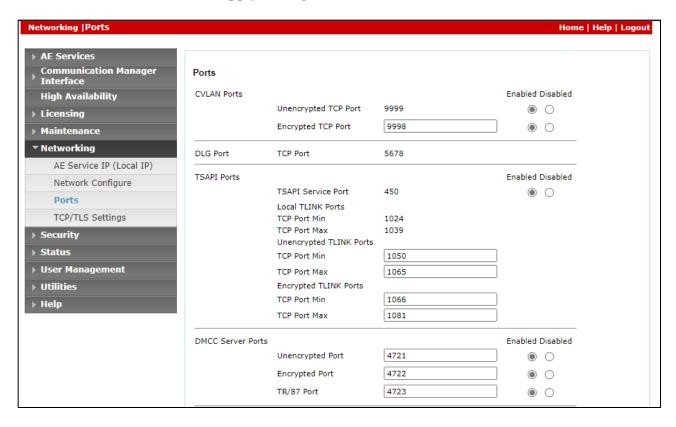


The **Edit CTI User** screen appears. Tick the **Unrestricted Access** box and **Apply Changes** at the bottom of the screen.



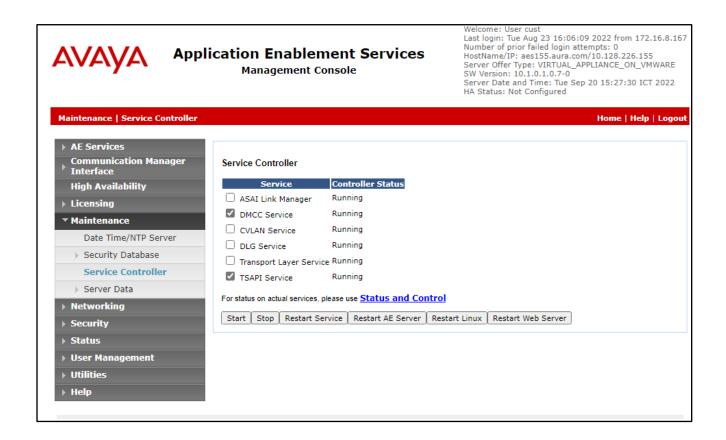
## 6.6. Display Port Information

Select **Networking**  $\rightarrow$  **Ports**, in the DMCC Server Ports section ensure that Unencrypted Port is Enabled and set to 4721. Click **Apply Changes** (not shown) when done.



#### 6.7. Restart Services

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



# 7. Configure Globitel SpeechLog Call Recording

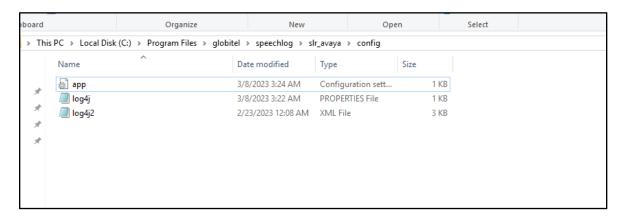
This section provides the procedures for configuring Globitel SpeechLog Call Recording. The procedures include following areas:

- Administer CTI server.
- Administer Virtual Ports.

The configuration of Globitel SpeechLog Call Recording is performed by Globitel engineers. The procedural steps are presented in these Application Notes are for informational purposes.

#### 7.1. Administer CTI Server

Navigate to the **C:\Program Files\globitel\speechlog\slr\_avaya\config** directory, and open the configuration file from the **SampleConfigFiles** directory to the current directory shown below.



Open the **app** file with the Notepad application, and configure as below:

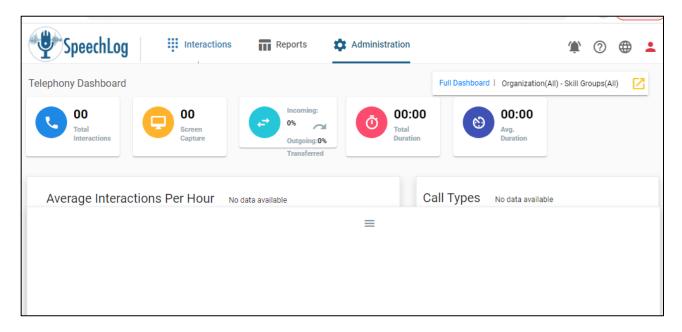
```
[app]
id = 1
type = tapi_mr
name = Globitel Speechlog Dmcc Service
host = IP address of Globitel Speechlog Call Recording server
storage_path = C:/globitel/calls/
[slc]
host = IP address of Globitel Speechlog Call Recording server
port = 55501
[cti]
host = Avaya AES IP Address
username = globitel
password = password for globitel user create in Section 6.4
cmapi_port = 4721 as per Section 6.6
tsapi port = 450 as per Section 6.6
tsapi_link = "AVAYA#CM121#CSTA#AES140"
```

## 7.2. Administer Register Devices

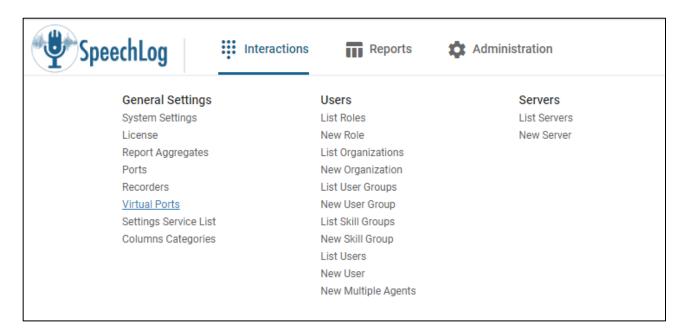
Access the Globitel SpeechLog Call Recording web-based interface by using the URL "http://ip-address" in an Internet browser window, where **ip-address** is the IP address of the Globitel SpeechLog Call Recording server. Log in using the appropriate credentials.



The screen below is displayed. Click on the **Administration** icon.



#### Select Virtual Ports.



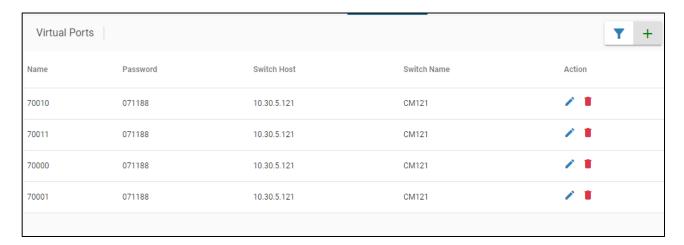
The **Virtual Ports** screen is displayed. Select + to add new **Virtual Ports**.



Input stations information that created in Section 5.2 and 5.3. Click Save.



For compliance testing these **Virtual Ports** created in Speechlog, which correspond to the two H.323 stations and the two SIP stations that were used and recorded.



# 8. Verification Steps

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

## 8.1. Verify Avaya Aura® Communication Manager

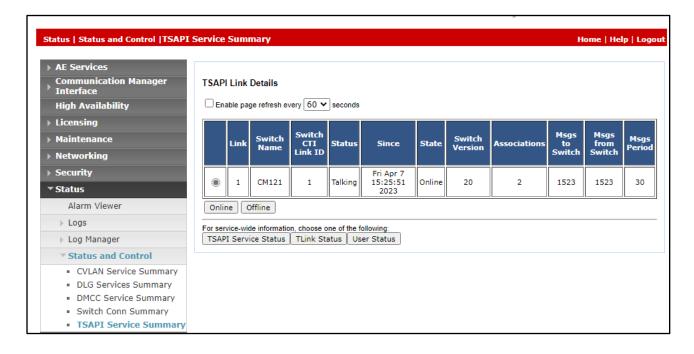
On Communication Manager, verify the status of the administered CTI link by using the **status aesvcs cti-link** command. Verify that the **Service State** is "established" for the CTI link number administered in **Section 5.1**, as shown below.

status aesvcs cti-link								
	AE SERVICES CTI LINK STATUS							
CTI Link	Version	Mnt Busy	AE Services Server	Service State	Msgs Sent	Msgs Rcvd		
1	12	no	aes140	established	1523	1523		

## 8.2. Verify Avaya Aura® Application Enablement Services

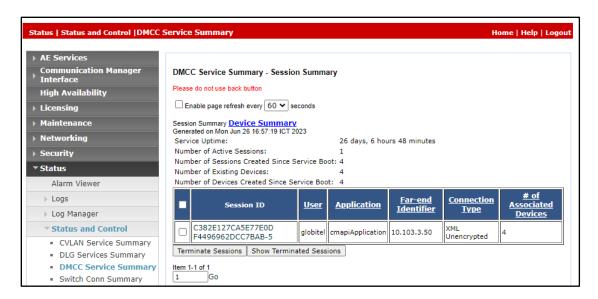
On Application Enablement Services, verify the status of the TSAPI link by selecting **Status Status and Control TSAPI Service Summary** from the left pane. The **TSAPI Link Details** screen is displayed.

Verify the **Status** is "Talking" for the TSAPI link administered in **Section 6.3**, and that the **Associations** column reflects the total number of monitored extensions from **Section 5.2** and **Section 5.3**.



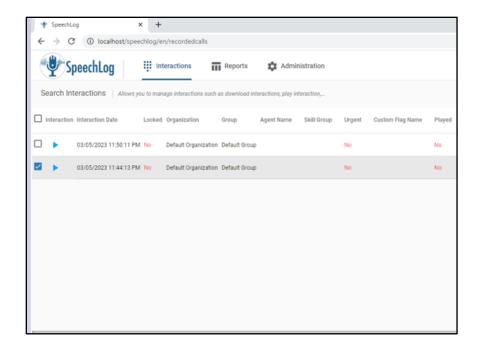
Verify the status of the DMCC link by selecting Status → Status and Control → DMCC Service Summary from the left pane. The DMCC Service Summary → Session Summary screen is displayed.

Verify the **User** column shows an active session with the Globitel user name from **Section 6.4**, and that the # of **Associated Devices** column reflects the total number of monitored extensions from **Section 5.2** and **Section 5.3**.

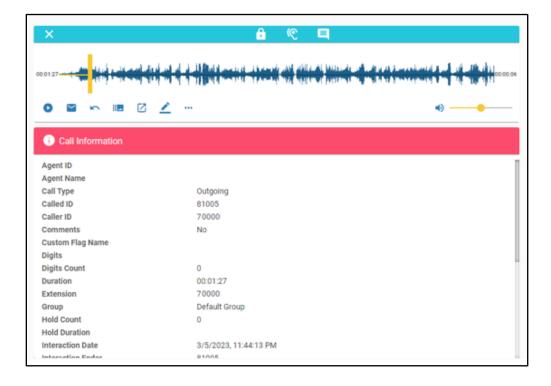


# 8.3. Verify Globitel SpeechLog Call Recording

Follow the procedures in **Section 7.2** to log in to the Globitel SpeechLog Call Recording 8.0 web-based interface. Select **Interaction**, it will show a listing of all recording entries.



Double click on the entry to listen to the playback. Verify that call recording is played back.



## 9. Conclusion

These Application Notes describe the configuration steps required for Globitel SpeechLog Call Recording 8.0 to successfully interoperate with Avaya Aura® Communication Manager and Avaya Aura® Application Enablement Services 10.1 using Multiple Registration. All feature and serviceability test cases were completed successfully.

## 10. Additional References

This section references the Avaya and Globitel SpeechLog Call Recording product documentation that are relevant to these Application Notes.

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

- 1. Administering Avaya Aura® Communication Manager, Release 10.1.x, Issue 1, Dec 2021
- 2. Administering Avaya Aura® Session Manager, Release 10.1.x, Issue 3, April 2022
- 3. Administering Avaya Aura® System Manager, Release 10.1.x, Issue 6, June 2022
- 4. Administering Avaya Aura® Application Enablement Services, Release 10.1.x, Issue 4, April 2022

Product Documentation for Globitel products may be found at <a href="https://www.globitel.com/">https://www.globitel.com/</a>

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