# Avaya Aura<sup>®</sup> Communication Manager configuration for BLU-103

Application Notes for Configuring BLU-103 VoIP Solution with Avaya Aura® Session Manager 6.3 and Avaya Aura® Communication Manager 6.3



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## 1. Introduction

BLU-103 VoIP device allows making and receiving phone calls over any Voiceover-IP (VoIP) system that adheres to the SIP (Session Initiation Protocol) standard.

The BLU-103 VoIP device integrates with Avaya SIP infrastructure consisting of Avaya Aura® Session Manager and Avaya Aura® Communication Manager as third-party SIP endpoints. These Application Notes describe the steps required to configure BLU-103 VoIP to work as Avaya third party SIP endpoints.

## 2. Interoperability Compliance

Interoperability compliance covers the following features and functionality:

- Successful registration of BLU-103 VoIP device with Session Manager
- Calls between BLU-103 VoIP device and Avaya SIP, H.323
- G.711, G.722 and G729/B codec support and negotiation, with and without media shuffling
- Basic features including phone display, mute/unmute, answer, hang up, music on hold, DTMF
- Proper system recovery of a BLU-103 VoIP telephone after removal and reconnection of LAN cable.

# 3. Configure Avaya Aura® Session Manager

This section provides the procedures for configuring Avaya Aura<sup>®</sup> Session Manager. The procedures include the following areas:

- Launch System Manager
- Administer users

### 3.1 Launch System Manager

- 3.1.1 Log on to the System Manager Web interface (https://← SERVER\_NAME →/ SMGR).
- 3.1.2 Enter your user name in the User ID field
- 3.1.3 Enter your password in the Password field.
- 3.1.4 Click Log On.

AVAYA	Avaya Aura * System Manager 6.3						
Home / Log On							
Log On							
Recommended access to Syst via FQDN. <u>Go to central login for Single</u> If IP address access is your o then note that authentication following cases:	tem Manager is <u>Sign-On</u> only option, n will fail in the	User ID: admin 😽 Password: 🗰 簧					
First time login with "ar Expired/Reset password" Use the "Change Password" this page to change the pass manually, and then login.	dmin" account rds hyperlink on sword	Supported Browsers: Internet Explorer 0.x, 9.x or 10.x or Firefox 15.0, 16.0 or 17.0.	Log On Cancel Change Password				

#### **FIGURE 3.1.4**

### Configure Avaya Aura® Session Manager

This section provides the procedures for configuring Avaya Aura® Session Manager.

### 3.2 Administer users

- 3.2.1 Go to User management page
- 3.2.2 Select Users  $\rightarrow$  Manage Users
- 3.2.3 In the User Management page, click New
- 3.2.4 In the General section, enter the user's last name and first name.
- 3.2.5 Enter a description in the Description field. This field is optional
- 3.2.6 For Login Name, enter "**extension@domain**", where extension is the BLU user extension
- 3.2.7 The Authentication Type should be Basic
- 3.2.8 Set Password

avaya	Avaya Aura® System Manager 6.3					L	Last Logged on at September 3, 2013 4:18 P Help   About   Change Password   Log adm		
							User Manage	ement × Home	
<sup>™</sup> User Management	Home	e / Users / U	lser Managei	ment / Manage I	Users				
Manage Users								Help ?	
Public Contacts	Us	er Mana	gement						
Shared Addresses									
System Presence ACLs	Us	ers							
	Vie	W Edit New	Duplicate	Delete More A	ctions •		A	dvanced Search 👁	
	7 Iter	ms Refresh S	how ALL					Filter: Enable	
		Last Name	First Name	Display Name	Login Name	SIP Handle	Last Login		
		ADVD	User1	ADVD User1	10070@harman.	com+10070			
		ADVD	User2	ADVD User2	10069@ harman.	com+10069			
			0100	AVAVA SIDS	10040@	com+10049			
		AVAYA	51P3	AVAIA, SIFS	10049@ harman.	COMPLOOP			

#### **FIGURE 3.2.3**

AVAYA	Avaya Aura® System	Manager 6.3		Last Logged on at Se Help   About   Ch	ptember 3, 2 ange Passw	013 4:18 PM vord   Log admi
				User Mana	gement *	Home
🕆 User Management	Home / Users / User Management / Ma	nage Users				
Manage Users						Help ?
Public Contacts	New User Profile			Commit & Continue	Commit	Cancel
Shared Addresses						
System Presence ACLs	Identity * Communication Profile	* Membership	Contacts			
	Identity 💌					
	* Last Name:	Line1				
	* First Name: 8	LU103				
	Middle Name:					
	Description:	* *				
	* Login Name:	0046@ harman.com				
	* Authentication Type:	Basic	Ŧ			
	Password:					
	Confirm Password:					



# Configure Avaya Aura® Session Manager

This section provides the procedures for configuring Avaya Aura® Session Manager.

- 3.2.9 Select communication profile
- 3.2.10 The Communication Profile Password must be administered. This is the Password that is used when logging in to the phone.
- 3.2.11 Click on the show/hide button for Communication Address.





#### 3.2.12 For each SIP handle

- Click New.
- Select Avaya SIP from the drop-down menu for Type if it is not set already.
- In the Fully Qualified Address field, enter the extension number.
- Click Add.





# Configure Avaya Aura® Session Manager

This section provides the procedures for configuring Avaya Aura® Session Manager.

#### 3.2.13 Assign the users to a Communication Manager station

#### Step1:

Scroll down to check and expand Session Manager Profile. For Primary Session Manager, Origination Application Sequence, Termination Application Sequence, and Home Location, select the values corresponding to the applicable Session Manager and Communication Manager. Retain the default values in the remaining fields. These settings are configured during the initial setup of Session Manager.

#### Step2:

Scroll down to check and expand CM Endpoint Profile. For System, select the value corresponding to the applicable Communication Manager. For Profile Type, select Endpoint. For Extension, enter the SIP user extension from Step 3.2.12. For Template, select "9630SIP\_DEFAULT\_CM\_6\_3". For Port, select "IP". Retain the default values in the remaining fields.

### Repeat Section 3.2 to add a user for BLU-103 VoIP Line2

You have now successfully added the SIP users to the Avaya Aura<sup>®</sup> Session Manager

🗑 Session Manager Profile 💌					
SIP Registration					
* Primary Sess	ion		Primary	Secondary	Maximum
Manag	er sml		7	0	7
Secondary Sess Manag	ion (None)	•			
Survivability Serv	(None)	-			
Max. Simultaneo Devic	es 1 💌				
Block New Registrat When Maxim Registrations Activ	ion 🗖 um 🗖 e?				
Application Sequences					
Origination Sequen	ce cm6-duplex-app-sed				
Termination Sequen	ce cm6-duplex-app-sed				
Call Routing Settings					
* Home Locati	on Location1				
Conference Factory S	et (None)				
🖉 CM Endpoint Profile 🛞					
* Syst	em CM6-duplex				
* Profile Ty	pe Endpoint				
Use Existing Endpoi	nts 🔲		_		
* Extens	ion Q 10046	Endpoint Editor			
* Templ	ate 9630SIP DEFAULT C	м 6 з 💌			
Set Ty	/pe 9630SIP				
Security Co	ode				
Р	ort QIP		]		
Voice Mail Num	ber		_		
Preferred Han	dle (None)				
Enhanced Callr-Info display for 1 phor	-line 📃 nes				
Delete Endpoint on Unassig	jn of 💟				

#### **FIGURE 3.2.13**

### 4. BLU-103 Configuration

Next, we'll focus on setting up the BLU-103 configuration to match the settings that were used in Aura® Communication Manager.

- Step 1 Open the Audio Architect software and discover the BLU-103 device
- **Step 2** Add the discovered BLU-103 to the venue
- **Step 3** Double-click on the device to view its internal audio configuration. Single-click on the "VoIP Input" Processing Object. Go to the properties grid.
- **Step 4** Under the "VoIP Network" category in the Properties select "DHCP", the VoIP interface will obtain an IP address automatically from a DHCP server. Otherwise, set "static" and manually enter an appropriate static IP Address, Subnet Mask, DNS and Gateway for the VoIP interface.
- Step 5 In the same tab, set "VLAN enabled" to "Checked", if your VoIP network uses a tagged VLAN. If the network uses an untagged VLAN or no VLAN, leave this value as "Disabled". If VLAN is enabled, enter the correct VLAN ID

### **BLU-103 VoIP Control/Status block**

BLU-103 VoIP Control		Avaya Aura <sup>®</sup> Communication Manager
User name	=	Extension number
Authentication name =		Login name
Password	=	Communication Profile Password

Properties	
Properties VolP Network VolP Network Time Line 1 Tones Line 1 Protocol Line 1 Timers Line 1 NAT Line 2 Tones Line 2 Protocol Line 2 To	
Ethernet Speed	Connection Type CHCP
IP Address	Subnet Mask
Gateway	DNS Primary
DNS Secondary	DNS Tertiary
VLAN Enabled	VLANID 0
VLAN Priority 0	STUN Server
STUN Port 3478	
The Ethernet Speed of the VolP card	



## **BLU-103** Configuration

This section focuses on setting up the BLU-103 configuration to match the settings that were used in Aura® Communication Manager.

- **Step 6** Under properties go to "Line 1 Protocol" tab
  - Enter the User Name and Authentication Name assigned to you when signing up for VoIP service
  - Enter the password to get authenticated to the VoIP service
  - Enter the Display Name in the field provided. This is also known as Caller ID.
  - Enter VoIP server address (i.e., Avaya Aura® Communication Manager's address) in the "Domain Name" field.
  - STUN and SRTP is disabled by default
  - Default Transport is Auto which uses UDP
- **Step 7** To register the Line 2 go to "Line 2 Protocol" and repeat the above step 6

#### Step 8 Go online in Audio Architect

- Step 9 The "Account Status" will be "Proxy Registered" once the registration is successful
- **Step 10** If the registration is unsuccessful it will display messages such as "Not configured", "AUTHENTICATION\_FAILURE" ....

Properties	operfies							
Properties VolP Netw	roperties VolP Network VolP Network Time Line 1 Teners Line 1 Protocol Line 1 Timers Line 2 Protocol Line 2 Timers Line 2 NAT							
User Name	3001	Authentication Name	3001					
Password	3001	Display Name	user 1001					
Domain Name	10.10.70.40							
STUN Enabled		Register with Domain	×					
Registration Life (s)	1600	Registration Retry (s)	x0					
Proxy Address		Praxy Part	0					
Signaling Port		SRTP Preference	Disabled					
Transport	Auto	RPort Enabled						
RTP Port Start	23456	RTP Port End						



### 5. Making calls from BLU-103

Follow the below steps to make a call

Placing a call from the DUT/AA is performed only if the user is registered to the network [Check the "account status" field]. Once the user is registered, follow the below steps to make a call.

- **Step 1** Double-click on the "VoIP Input" Processing Object to open its default panel.
- **Step 2** Enter the registered number you wish to dial either from Line 1 or Line 2 (use the tabs at the top of the panel to switch between the two lines)
- **Step 3** Click on the call button [Visible in green]
- **Step 4** Once the conversation is done press the disconnect button [Visible in red]

	VoIP LINE	1		VoIP LINE 2	
			SPEE	D DIAL NAME SPEED DIAL I	NUMBER
1	2	3			
4	5	6			
7	8	9			
*	0	#	9		
PAUSE		DELETE			
REDIAL		BACKSPACE			
•	HOLD	<b>↓</b> »)			
0.00dBu OLIN -6 12	12	+20 0.00dBu	0.00dBu	AUTO ANSWER OFF	LINESTATUS 🥥
66 0	ee ee		4- <b></b> -4 0- <b></b> -4 -4- <b></b> -4	INCC	DMING CALL
	**************************************	-1212 -20 -2020 -30 RING	2 -1212 -2020 DTMF	Reject Call voip input ca	rd 📑