

Avaya Solution & Interoperability Test Lab

Application Notes for Phybridge UniPhyer with Avaya AuraTM Communication Manager and Avaya AuraTM SIP Enablement Services – Issue 1.0

Abstract

These Application Notes describe the configuration steps required for the Phybridge UniPhyer LAN appliance to interoperate with Avaya AuraTM Communication Manager and Avaya AuraTM SIP Enablement Services. In the compliance testing, Phybridge UniPhyer leveraged the existing single-pair telephony wiring to provide dedicated Ethernet voice path and Power over Ethernet to Avaya IP Telephones connected to Avaya AuraTM Communication Manager and Avaya AuraTM SIP Enablement Services.

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 the Phybridge UniPhyer LAN appliance to interoperate with Avaya AuraTM Communication Manager and Avaya AuraTM SIP Enablement Services.

In the compliance testing, three analog telephones on Avaya AuraTM Communication Manager were replaced with Avaya IP Telephones. The existing RJ11 cabling for the analog connections were reused by Phybridge UniPhyer to connect to the new Avaya IP Telephones, and to provide dedicated Ethernet voice path and Power over Ethernet.

1.1. Interoperability Compliance Testing

The interoperability compliance test included feature and serviceability testing.

The feature testing included firmware download, registration, audio codec, media shuffling, basic call, hold/reconnect, conference, transfer, display, call forwarding, DTMF, button activation/deactivation, feature access code activation/deactivation, and message waiting lamp scenarios.

The serviceability testing focused on verifying the ability of Phybridge UniPhyer to recover from adverse conditions, such as disconnecting and reconnecting the Ethernet cables to Phybridge UniPhyer and to Avaya IP Telephones.

1.2. Support

Technical support on Phybridge UniPhyer can be obtained through the following:

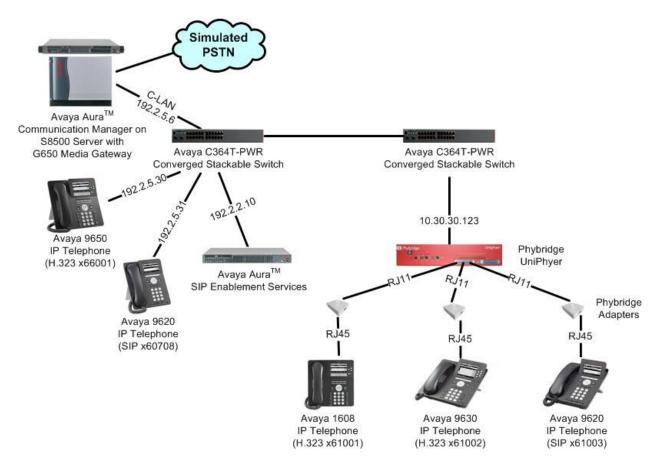
- Phone: (888) 901-3633
- Email: <u>techsupport@phybridge.com</u>

2. Reference Configuration

In the test configuration shown below, three analog telephones on Communication Manager were replaced with Avaya IP Telephones by leveraging the existing RJ11 cabling. For each station user, one end of the RJ11 cable was changed to connect to Phybridge UniPhyer instead of Communication Manager, and the other end of the RJ11 cable was connected to a Phybridge Adapter. For each Phybridge Adapter, there was a RJ45 cable connection to an Avaya IP Telephone.

In the compliance testing, two of the Avaya IP Telephones were provisioned with H.323 and registered to Communication Manager, and the other Avaya IP Telephone was provisioned with SIP and registered to Avaya AuraTM SIP Enablement Services.

The Phybridge UniPhyer provided power to the connected Avaya IP Telephones, and acted as a straight pass through and transparent to these Avaya IP Telephones in terms of the telephones' network settings.



3. Equipment and Software Validated

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

Equipment	Software
Avaya Aura TM Communication Manager on Avaya S8500 Server	R015x.02.0.947.3
Avaya G650 Media GatewayTN799DP C-LAN Circuit Pack	HW01 FW017
Avaya Aura TM SIP Enablement Services	SES-5.2.0.0-947.3a
Avaya 1608 IP Telephone (H.323)	1.2
Avaya 9600 Series IP Telephones (H.323)	3.0
Avaya 9620 IP Telephones (SIP)	2.4.1
Phybridge UniPhyer	0.78B03
Phybridge Adapters	007-001 Rev 2

4. Configure Avaya Aura[™] Communication Manager

This section provides the procedures for configuring Avaya AuraTM Communication Manager. The procedures fall into the following areas:

- Pre-configuration
- Verify Communication Manager license
- Modify stations
- Administer off-PBX

The detailed administration of basic connectivity between Communication Manager and SIP Enablement Services is not the focus of these Application Notes and will not be described.

4.1. Pre-Configuration

Log in to the System Access Terminal (SAT). Prior to start of test, three analog stations were connected to Communication Manager, as shown below.

```
list station 61002 count 3
                              STATIONS
                                           Room/ Cv1/ COR/ Cable/
Ext/
Hunt-to
Ext/
          Port/ Name/
                                    Move Data Ext
            Type Surv GK NN
                                                        Cv2 COS Jack
          01A0901 Phybridge User #1
61001
                                                             1
            2500
                                                              1
                                      no
61002
            01A0902 Phybridge User #2
                                                             1
            2500
                                                              1
                                      no
61003
            01A0903 Phybridge User #3
                                                             1
             2500
                                      no
                                                              1
```

4.2. Verify Communication Manager License

Use the "display system-parameters customer-options" command to verify that there is sufficient capacity for SIP stations by comparing the **Maximum Off-PBX Telephones - OPS** field value with the corresponding value in the **USED** column. In the compliance testing, one analog station was changed over to a SIP station.

```
display system-parameters customer-options
                                                                   1 of 11
                                                              Page
                              OPTIONAL FEATURES
    G3 Version: V15
                                               Software Package: Standard
      Location: 1
                                            RFA System ID (SID): 1
      Platform: 12
                                            RFA Module ID (MID): 1
                                                            USED
                              Platform Maximum Ports: 3200 234
                                   Maximum Stations: 2400 173
                            Maximum XMOBILE Stations: 0
                                                            0
                   Maximum Off-PBX Telephones - EC500: 0
                                                            0
                   Maximum Off-PBX Telephones - OPS: 100
                                                            6
                   Maximum Off-PBX Telephones - PBFMC: 0
                                                            0
```

Navigate to **Page 2**, and verify that there is sufficient capacity for IP stations by comparing the **Maximum Concurrently Registered IP Stations** field values. In the compliance testing, two analog stations were changed over to H.323 stations.

display system-parameters customer-options OPTIONAL FEATURES		Page	2 of	11
IP PORT CAPACITIES		USED		
Maximum Administered H.323 Trunks:	100	6		
Maximum Concurrently Registered IP Stations:	18000	2		
Maximum Administered Remote Office Trunks:	8000	0		
Maximum Concurrently Registered Remote Office Stations:	18000	0		
Maximum Concurrently Registered IP eCons:	10	0		
Max Concur Registered Unauthenticated H.323 Stations:	10	0		
Maximum Video Capable H.323 Stations:	100	0		
Maximum Video Capable IP Softphones:	100	0		
Maximum Administered SIP Trunks:	10	10		
Maximum Administered Ad-hoc Video Conferencing Ports:	0	0		
Maximum Number of DS1 Boards with Echo Cancellation:	0	0		
Maximum TN2501 VAL Boards:	10	0		
Maximum Media Gateway VAL Sources:	10	0		

4.3. Modify Stations

After installation of Phybridge UniPhyer, each analog telephone was replaced with an Avaya IP Telephone, and the RJ11 cable was reconnected as described in **Section 2**. This section modifies the station type for each user to match the new Avaya IP Telephone, and allows the user to retain the same extension number.

Change the station type of an existing analog station by using the command "change station n", where "n" is the existing extension number. For **Type**, enter the applicable IP station type, in this case "1608", and the **Port** field will be populated automatically. Enter a desired **Security Code**.

change station 61001			Pag	ge 1 of	4
		STATION			
Extension: 61001		Lock Messages?	n	BCC:	0
Type: 1608		Security Code:		TN:	•
Port: IP		Coverage Path 1:		COR:	
Name: Phybridge User	#1	Coverage Path 2: Hunt-to Station:		COS:	1
STATION OPTIONS					
		Time of Day L	ock Table:		
Loss Group:	19	Personalized Ringin	g Pattern:	1	
		Message	e Lamp Ext:	61001	
Speakerphone:	2-way	Mute Butto	on Enabled?	У	
Display Language:	english				
Survivable GK Node Name:					
Survivable COR:		Media Co	omplex Ext:		
Survivable Trunk Dest?	У	IP	SoftPhone?	n	

Repeat this section to modify the station type for all applicable analog and/or digital stations. In the compliance testing, three analog stations were changed over to IP.

list station	61001 co	unt 3				
		STATION	S			
Ext/ Hunt-to	Port/ Type	Name/ Surv GK NN	Move	Room/ Data Ext		COR/ Cable/ COS Jack
61001	S00001 1608	Phybridge User #1	no			1
61002	S00002 9630	Phybridge User #2	no		1	1
61003	s00003 9620	Phybridge User #3	no			1 1

4.4. Administer Off-PBX

For each SIP station, specify that calls to the station be routed to SIP Enablement Services by using the "change off-pbx-telephone station-mapping n" command, where "n" is the station extension. Enter the following values for the specified fields, and retain the default values for the remaining fields. In the compliance testing, station "61003" was configured as a SIP station.

- Application: Enter "OPS" to indicate off-PBX station.
- **Phone Number:** Same digits from the **Station Extension** field.
- Trunk Selection: The trunk group to reach SIP Enablement Services, in this case "5".
- **Config Set:** An existing configuration set to be used for the off-PBX call treatment.

change off-pbx	-		ing 61003 BX TELEPHONE INT:		Page 1 d	of 3
Station Extension 61003	Application OPS	Dial CC Prefix -	Phone Number 61003	Trunk Selection 5	Config Set 1	Dual Mode
		-				

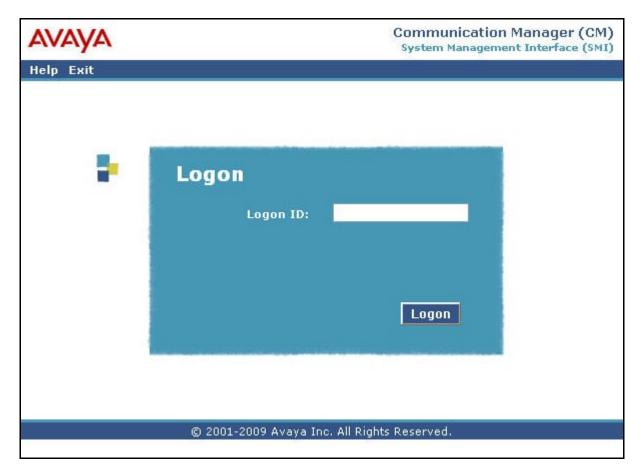
5. Configure Avaya Aura[™] SIP Enablement Services

This section provides the procedures for configuring Avaya AuraTM SIP Enablement Services. The procedures include the following areas:

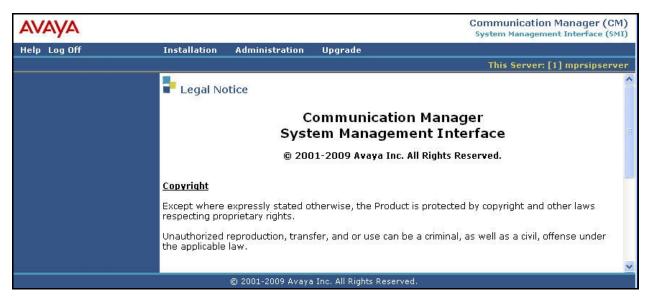
- Launch administration interface
- Administer users

5.1. Launch Administration Interface

Access the SIP Enablement Services web interface by using the URL "http://ip-address/admin" in an Internet browser window, where "ip-address" is the IP address of the SIP Enablement Services server. Log in with the appropriate credentials.



In the subsequent screen, select **Administration > SIP Enablement Services** from the top menu.



The Top screen is displayed next.

AVAYA		Integrated Ma SIP Server M	
Help Exit			1] mprsipserver
Top ■ Users Address Map Priorities	Top Manage Users	Add and delete Users.	ei
 Adjunct Systems Aggregator 	Manage Address Map Priorities	Adjust Address Map Priorities.	÷3
 Certificate Management Conferences 	Manage Adjunct Systems	Add and delete Adjunct Systems.	
Emergency Contacts Export/Import to ProVision	Manage Event Aggregators	Add/Delete Event Aggregators.	-
# Hosts	Certificate Management	Manage Certificates.	
IM logs Communication Manager Servers	Manage Conferencing	Add and delete Conference Extensions.	
 Communication Manager Extensions 	Manage Emergency Contacts	Add and delete Emergency Contacts.	
 Server Configuration SIP Phone Settings 	Export Import to ProVision	Export and import data using ProVision on this host.	
Survivable Call Processors	Manage Hosts	Add and delete Hosts.	•8
System Status	IM logs	Download IM Logs.	
 Trace Logger Trusted Hosts 	Manage Communication Manager Servers	Add and delete Communication Manager Servers.	-

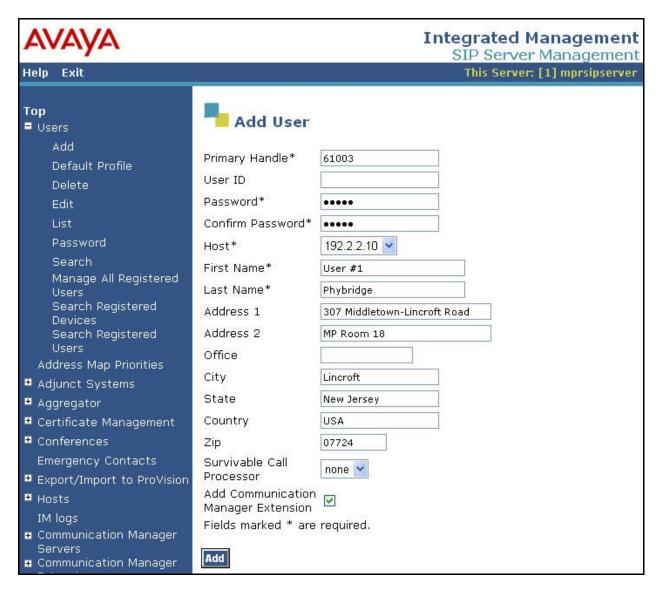
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5.2. Administer Users

Select Users > Add from the left pane to display the Add User screen. Enter the following values for the specified fields, and retain the default values for the remaining fields.

- **Primary Handle:** The extension of the SIP station from **Section 4.4**.
- **Password:** Enter a desired password.
- **Confirm Password:** Re-enter the same password.
- Host: Select the applicable host.
- First Name: A descriptive first name.
- Last Name: A descriptive last name.

Check the Add Communication Manager Extension field, and click Add.





The Continue screen is displayed next. Click Continue.

The Add Communication Manager Extension screen is displayed. This screen is used to associate a user with an extension on Communication Manager. For Extension, enter the SIP station extension from Section 4.4. Select the appropriate Communication Manager Server, and click Add.

AVAYA	Integrated Management SIP Server Management
Help Exit	This Server: [1] mprsipserver
Top ■ Users Add Default Profile Delete Edit List Password Search Manage All Registered Users	Add Communication Manager Extension Add Communication Manager extension for user 61003. Extension 61003 Communication Manager Clan-1 Server Fields marked * are required.

6. Configure Phybridge UniPhyer

This section provides the procedures for configuring Phybridge UniPhyer. The procedures fall into the following areas:

- Launch web interface
- Administer board IP
- Save configuration

6.1. Launch Web Interface

Access the Phybridge UniPhyer web interface by using the URL "http://ip-address" in an Internet browser window, where "ip-address" is the IP address of Phybridge UniPhyer. Note that the default IP address of the Phybridge UniPhyer management port is "192.168.1.1". The **Web Interface Login** screen is displayed as shown below. Log in using the appropriate credentials.

Phybridge
UniPhyer
Web Interface Login
Username: Password: Sign in
 Level 1:SuperUser, R/W Management all Level 2:Engineer, R/W (Disabled from User Account) Level 3:Guest, Read only

6.2. Administer Board IP

In the subsequent screen, select **System > Board IP Setup** to display the **Board IP Setup** screen. Modify the **IP Address** and **Subnet Mask** fields in the **GBE (In Band)** and **MGMT** (**Out Band)** sections to match the network configuration. Click **Modify**, followed by **RESTART**.

Note that the **MGMT (Out Band)** configuration is optional, and needs to be on a different subnet from the **GBE (In Band)** if used.

Phybridge	U	niPhyer		
• System		Board	d IP Setup	
System Info Board IP Setup	Modify	RESTART		
Ethernet Port Service		Address	Managemen	t
ADSL Port Service		GBE (In Band)		MGMT (Out Band)
CLI Setup Cluster Setup	IP Address	10 . 30 . 30 . 123	IP Address	192 168 1 1 1
System Inventory System Contact Info	Subnet Mask	255 . 255 . 255 . 0	Subnet Mask	255 . 255 . 255 . 0
SNTP IP Routes	NO Limit VID		DHCP Client	Disable DHCP Client 💌
User Administration Duplicator	Limit VID		DHCP Timeout	60
■ Bridge	Priority	0 🕶	DHCP Lease	4294967295
	HTTP Port	MGMT Speed	Remote IP	System Name
SNMP	80	Auto Negotiate	192.168.1.10	UniPhyer
Maintenance	[System In	nventory]		
	Modify th	e configuration may cause the co	onnection los	S

6.3. Save Configuration

Select Maintenance > Database to display the Database Configuration screen. For the DB Config Select field, select "(D)Save Running Config to Flash(System Config)".

Phybridge	UniPhyer	
≠ System ● Bridge	Database Configuration	
 ADSL Traffic SNMP Maintenance SYS Log Server Database Firmware Update Boot Code Update ATM Loopbacks Fault Management Performance Monitoring 	DB Config Select [Select] (A)Import File (Write Download Config To FLASH) (B)Import File (Load Remote Config to Running Config) (C)Export File (Put Running Config To Remote TFTP Server) (D)Save Running Config to Flash(System Config) (E)Reload FLASH(System Config) to Running Config (F)Restore Factory Default (G)Flash Boot Point Configuration Select	

The screen is updated as shown below. Click **Write_Running** to save the modified configuration to flash.

Phybridge	UniPhyer
 System Bridge ADSL Traffic SNMP Maintenance SYS Log Server Database Firmware Update Boot Code Update ATM Loopbacks Fault Management Performance Monitoring 	Database Configuration DB Config Select: (D)Save Running Config to Flash(System Config) Write flash at: Partition1 Write flash at: Partition1

7. General Test Approach and Test Results

All tests were performed manually. The focus was on verifying that the Avaya H.323 and SIP Telephones can function seamlessly, when connected via Phybridge UniPhyer.

All tests were executed and passed.

8. Verification Steps

This section provides the tests that can be performed to verify proper configuration of Avaya AuraTM Communication Manager, Avaya AuraTM SIP Enablement Services, and Phybridge UniPhyer.

8.1. Verify Avaya Aura[™] Communication Manager

Use the "list registered-ip-stations" command to verify that all H.323 stations connected via Phybridge UniPhyer from **Section 4.3** registered successfully with Communication Manager, as shown below.

```
      REGISTERED IP STATIONS

      Station Ext
or Orig Port
      Set Type/
Net Rgn
      Prod ID/
Release
      TCP Station IP Address/
Skt Gatekeeper IP Address

      61001
      1608
      IP_Phone
1
      y
      10.30.30.101
192.2.5.6

      61002
      9630
      IP_Phone
1
      y
      10.30.30.102
192.2.5.6

      66001
      9650
      IP_Phone
1
      y
      192.2.5.101
192.2.5.6
```

8.2. Verify Avaya Aura[™] SIP Enablement Services

From the SIP Enablement Services web interface, verify the registration status of the SIP stations by selecting Users > Search Registered Users from the left pane. Verify that all SIP stations from Section 4.4 are listed as registered users.



8.3. Verify Phybridge UniPhyer

From the Phybridge UniPhyer web interface, select **System > ADSL Port Service**. The **ADSL Port Service** screen is displayed. Verify that the **Current Status** for all physically connected voice ports is in the **ON** state, as shown below.

E System				ADSL Port S	Service		
System Info Board IP Setup							
Ethernet Port Service	Admin ON		vice Profile	2 Spectru	m Profile 2	TCA Profile	2
ADSL Port Service CLI Setup		odify Profile rap	qe from 1 to 1	20			
Cluster Setup		um Profile ra	ange from 1 to				
System Inventory System Contact Info	Port 01~12		_				
SNTP	Select	Port	Admin Status	Current Status	Service Profile	Spectrum Profile	TCA Profile
IP Routes User Administration	۲	1	ON	ON	2	2	2
Duplicator	0	2	ON	ON	2	2	2
Bridge ADSL	0	3	ON	ON	2	2	2
Traffic	0	4	ON	OFF	2	2	2
SNMP	0	5	ON	OFF	2	2	2
# Maintenance	0	6	ON	OFF	2	2	2
	0	7	ON	OFF	2	2	2
	0	8	ON	OFF	2	2	2
	0	9	ON	OFF	2	2	2
	0	10	ON	OFF	2	2	2
	0	11	ON	OFF	2	2	2
	and the second sec	12	ON	OFF	2	2	2

9. Conclusion

These Application Notes describe the configuration steps required for Phybridge UniPhyer to successfully interoperate with Avaya AuraTM Communication Manager and Avaya AuraTM SIP Enablement Services via Avaya IP Telephones.

10. Additional References

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

- 1. Administering Avaya AuraTM Communication Manager, Document 03-300509, Issue 5.0, Release 5.2, May 2009, available at <u>http://support.avaya.com.</u>
- **2.** *SIP Support in Avaya Aura*TM *Communication Manager*, Document 555-245-206, Issue 9, May 2009, available at <u>http://support.avaya.com</u>.
- **3.** Installing, Administering, Maintaining, & Troubleshooting Avaya AuraTM SIP Enablement Services, Document 03-600768, 7.0, May 2009, available at <u>http://support.avaya.com.</u>
- **4.** Avaya-Phybridge DevConnect Partner Solutions Guide, Release 1.0, available at <u>http://www.phybridge.com</u>.
- 5. *Phybridge UniPhyer Installation Manual*, Release 1.0, available at <u>http://www.phybridge.com</u>.
- 6. *Phybridge UniPhyer Web Configuration Tool Guide*, Release 1.0, available at <u>http://www.phybridge.com</u>.

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