Ribbon ATA EdgeMarc 300 R16.2 Interop with Zoom : Interoperability Guide



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Interoperable Vendors



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Document Overview

This document outlines the configuration best practices for the Ribbon solution covering the EdgeMarc 300 series ATA (Analog Telephone Adaptor) when deployed with Zoom.

The Ribbon EdgeMarc 300 is a low-density analog gateway or ATA device that gives small businesses, SOHOs and branch offices with analog voice infrastructures an easy, cost-effective way to capitalize on Voice over Internet Protocol (VoIP) services. The EdgeMarc 300 Series offers a solution, providing support for integrating analog endpoints and the Public Switched Telephone Network (PSTN) and support for all Session Initiation Protocol (SIP) calls. The EdgeMarc 300 supports any-to-any connectivity between analog and SIP devices, enabling branch offices to rapidly migrate analog phones onto SIP-based networks and communicate seamlessly.

The interoperability compliance testing focuses on verifying inbound and outbound call flows between Ribbon EdgeMarc ATA & Zoom Cloud.

This guide contains the following configuration sections:

- Section A: Ribbon EdgeMarc Configuration
 - Captures general EdgeMarc 300 configurations for provisioning with Zoom.
- Section B: Zoom Configuration
 - Captures the Zoom configuration.
- All basic calls, along with the supplementary features like call hold, call transfer, and conference can be tested with configurations from Section A and Section B.
- Advanced supplementary features can be configured on Zoom as mentioned in Supplementary Services Configuration on Zoom. These cover:
 - Auto Receptionist
 - Call Flip
 - Shared Line Appearance (SLA) or Call Delegation
 - Shared Line Group (SLG)

Non-Goals

It is not the goal of this guide to provide detailed configurations that will meet the requirements of every customer. Use this guide as a starting point and build the ATA configurations in consultation with network design and deployment engineers.

Audience

This is a technical document intended for telecommunications engineers with the purpose of configuring the Ribbon ATA.

To perform this interop, you need

- to use graphical user interface (GUI) or command line interface (CLI) of the Ribbon product.
- to understand the basic concepts of TCP/UDP/TLS and IP/Routing.

• to have SIP/RTP/SRTP to complete the configuration and for troubleshooting.

Note

This configuration guide is offered as a convenience to Ribbon customers. The specifications and information regarding the product in this guide are subject to change without notice. All statements, information, and recommendations in this guide are believed to be accurate but are presented without warranty of any kind, express or implied, and are provided "AS IS". Users must take full responsibility for the application of the specifications and information in this guide.

Prerequisites

The following aspects are required before proceeding with the interop:

- Ribbon EdgeMarc 300 series
- Public IP Addresses
- Zoom Go account a special type of Zoom account that has the option to provision the ATA device.
- TLS Certificates for Ribbon EdgeMarc 300 series
 - Refer to TLS Configuration between Ribbon EdgeMarc and Zoom

Product and Device Details

The sample configuration in this document uses the following equipment and software:

 Table 1: Requirements

	Equipment	Software Version
Ribbon Communications	Ribbon EdgeMarc 304	V16.2.0.sm.EM-26519.1
Zoom	Zoom Desktop app	5.7.7 (1105)
	Zoom Mobile app	5.8.1 (2403)
Third-party Phones	Beetel Analog Phone	NA

Note

- The Ribbon EdgeMarc 300 portfolio includes EdgeMarc 302, EdgeMarc 304. Hence, this configuration guide is valid for all these devices.
- Zoom Desktop app version is 5.7.7 (1105) or later.
- Zoom Mobile app version is 5.8.1 (2403) or later.

Network Topology Diagram

This section covers the Ribbon EdgeMarc deployment topology and the Interoperability Test Lab Topology.

Ribbon EdgeMarc Deployment Topology

Figure 1: Ribbon EdgeMarc Deployment Topology



Interoperability Test Lab Topology

The following lab topology diagram shows connectivity between Zoom and Ribbon EdgeMarc ATA.



Figure 2: Interoperability Test Lab Topology

Section A: EdgeMarc Configuration

The following EdgeMarc configurations are included in this section:

- 1. Connectivity
- 2. Network
- 3. SIP UA

Connectivity

Below is an image of the EdgeMarc back panel:

Figure 3: EdgeMarc Back Panel



The following table provides details of the EdgeMarc 304 interface/port:

Figure 4: EdgeMarc 304 Interfaces

	EdgeMarc 304
Ports	
LAN 1 Gb/s Ethernet (RJ-45)	4
Phone/FXS (RJ-11)	4
Micro SD (SDXC) slot	1
Console (RJ-45)	1

The current test bed setup uses the following ports:

- LAN Port RJ45 "LAN 1" port is connected to Public network.
- Phone Port RJ11 "Phone 1" and "Phone 2" port connects to a PSTN (Analog) phones.

(I) WAN and SFP WAN ports of EdgeMarc 304 are disabled in the ATA (Analog Telephone Adapter) mode.

Network

0

To configure the network interface, login to the EdgeMarc as root user and click Network.

Figure 5: EdgeMarc Network Interface

noddin 🖏	Network	Help
Configuration Menu + Admin - Network • VLAN • 802.1X Supplicant • Switch Ports • Static Routes • Static Routes • Network Information • Network Restart • Network Test Tools + Users • Certificates • HTTPS Configuration + SIP UA	Network Interface Settings Enable DHCP: IP Address: Subnet Mask: Default Gateway: Use DNS addresses from provider: Primary DNS Server: Secondary DNS Server: Enable LLDP-MED: Enable VLAN: Default VLAN ID: VLAN Priority:	115.110. . 255.255. . 115.110. . 8.8.8.8 . . . 1 . 5 .

SIP UA

To configure SIP UA:

1. Navigate to SIP UA.

noddin 🛟	FXS/Phone Port Settings - Bas	ic	<u>Help</u>
~	SIP UA allows voice call from Analog port to IP	or PSTN	
Configuration Menu	Global configuration:		
+ <u>Admin</u> + <u>Network</u> + <u>Users</u> • <u>Certificates</u> • <u>HTTPS Configuration</u> - <u>SIP UA</u> • <u>Advanced</u> • <u>Fax</u> • Distinctive Ring	Enable SIPUA: Use SIP Username for SIP authentication: Codec Preference: Use Preferred codec only: Use REFER for transfer: Register with proxy:	 ✓ G.711 ulaw ✓ ✓ ✓ ✓ ✓ 	

A Port 1 and Port 2 configuration details can be retrieved once the EM 300 ATA is added to the Zoom portal.

To add EM 300 ATA to the Zoom portal, navigate to Adding MAC address.

2. Assign Port 1 and Port 2 with the PSTN number as follows:

Figure 7: Port Level Basic Config

Port Level Basic Configuration		
Port 1 Configuration: (Registered)		
Hook state: On-hook		
SIP Display name:	10000587.zoom.us	
SIP Username:	27894760291355532685	
SIP Authentication name:	258417410964	
Password:	is set	
Edit Password:		
Password:		
Confirm Password:		
Port 2 Configuration: (Registered)		
Hook state: On-hook		
SIP Display name:	10000587.zoom.us	
SIP Username:	71983719222258799004	
SIP Authentication name:	107768177695	
Password:	is set	
Edit Password:		
Password:		
Confirm Password:		

3. Navigate to $\ensuremath{\text{SIP UA}}$ Advanced and configure as shown below:

Figure 8: Advanced Config

noddir 🛟	FXS/Phone Port Settings - Adv	vanced	
*	This Page allows advance configuration of FXS/	Phone ports	
Configuration Menu	Enable SIPUA:		
+ <u>Admin</u> + <u>Network</u> + Users	Global configuration: Reset to Defaults		
<u>Certificates</u> <u>HTTPS Configuration</u>	SIPUA bind Port: Conference URI:	5061	
<u>Advanced</u> Fax	Domain:		
• <u>Distinctive Ring</u> • <u>SIP Proxies</u>	RTP Max Port:	20150	
	Termination Impedance: Local TimeZone:	600r ohms V UTC V	
	Dialed in prefix(Incoming from IP network):	000	
	Ptime:		
	Inter Digit Delay timer(in seconds): Enable Access Code (#):		
	Enable Call Waiting:		
	Internal Call Ring:	Ring-0 V	
	External Call Ring: Dial-Completion Pattern:	Ring-0 V]
	VAD Enable: Country:	On v USA v	

Port Level Advanced Configuration		
Port 1 Configuration:	*	
Codec Preference:	Global Setting V	
Use Preferred codec only:		
Domain:	10000587.zoom.us	
Outbound Proxy Server IP:		
Outbound Proxy Server Port:	5091	
Outbound proxy Server Transport:	TLS V	
Enable SRTP:	Global Setting 🗸	
TX gain to analog device:	ODB V	
RX Gain from analog device:	ODB V	
Enable Access Code (#):		
Enable Call Waiting:		
Enable Call Waiting CallerID:		
VAD Enable:	Global Setting V	
Hotline number:		
Restrict CallerID:		
Disable Flash Hook:		

Port 2 Configuration:	
Codec Preference:	Global Setting 🖌
Use Preferred codec only:	
Domain:	10000587.zoom.us
Outbound Proxy Server IP:	
Outbound Proxy Server Port:	5091
Outbound proxy Server Transport:	TLS V
Enable SRTP:	On 🗸
TX gain to analog device:	0DB 🗸
RX Gain from analog device:	0DB 🗸
Enable Access Code (#):	
Enable Call Waiting:	
Enable Call Waiting CallerID:	
VAD Enable:	Global Setting V
Hotline number:	
Restrict CallerID:	
Disable Flash Hook:	

4. Navigate to SIP UA SIP Proxies.

Figure 9: SIP Proxies

noddin 🛟	SIP Proxy	/ Settings		Hei
-	This page allov	vs configuration of SIP Proxy Settings.		
Configuration Menu		List of SIP Servers		Delete All
+ Admin	Priority	SIP Server Address	Port	Action
+ <u>Network</u> + Users	0	gosip0h.sc.zoom.us	5091	⊗↑↓
<u>Certificates</u> HTTPS Configuration				Add
- <u>SIP UA</u>	Outbound proxy Server Transport: TLS Control TLS Contro			
• <u>Advanced</u>				
• <u>Fax</u> • <u>Distinctive Ring</u> • SIP Provies				TLS.
	SIPUA TLS se	ettings		
	Port:	5091		
	TLS Protocol:	TLSv1.2 V	_	
	Ciphers String	TLSv1.2+HIGH:!eNULL:!aN	ι	
	Interface:	Certificate: Default 🗸 🛛	Policy: Require an	d Verify 🗸 🗸
	Submit Rese	Apply Later		

Section B: Zoom Configuration

Login to Zoom Go account Web portal at https://go.zoom.us/.

This section describes the following Zoom configurations:

- 1. Adding MAC address to Zoom.
- 2. TLS Configuration between Ribbon EdgeMarc and Zoom
- 3. Configuring supplementary services configuration on Zoom

Adding MAC Address to Zoom

- 1. Navigate to Phone Systems Management > Phones & Devices > Deskphone.
- 2. Select Add Desk Phone and add the MAC address of the Ribbon EdgeMarc 304.
- 3. Assign the Zoom Users as shown below and click $\ensuremath{\textbf{Save}}$.

Figure 10: Add MAC Address



Figure 11: Add Desk Phone

Add Device	
Display Name	EdgeMarc 304
Description (Optional)	
MAC Address	54:39:68:1B:27:4D
Device Type	Other v
	This device type supports up to 1 assignee.
Assigned to	ankit shukla × Assign Ext. 803
	Save Cancel

4. Provision the User as follows:

Figure 12: Actions

No description Profile Site Main Site Assigned to ankit s IP Address 115.110. Device Type Other Firmware Version MAC Address 54-5 Provision Template Uns	e (Main Site) hukla × 170.214
Profile Site Main Sit Assigned to ankit s IP Address 115.110. Device Type Other Firmware Version MAC Address 54-3 Provision Template Uns	e (Main Site) hukla × 03 × 170.214
Site Main Sit Assigned to ankit s Ext. 80 IP Address 115.110. Device Type Other Firmware Version MAC Address 54-3 Provision Template Uns	e (Main Site) hukla × 03 × 170.214
Assigned to ankit s Ext. 80 IP Address 115.110. Device Type Other Firmware Version MAC Address 54-3 Provision Template Uns	hukla × 03 × 170.214
IP Address 115.110. Device Type Other Firmware Version MAC Address 54-3 Provision Template Uns	170.214
Device Type Other Firmware Version MAC Address 54-3 Provision Template Uns	
Firmware VersionMAC Address54-3Provision TemplateUns	
MAC Address 54-3 Provision Template Uns	
Provision Template Uns	39-68-1b-27-4d Edit
	upported 2
Status Onl	
Actions V Remove	ine

Click Actions > Provision, as shown above:

Figure 13: Provisioning

D · · · ·		
Provisioning		
MAC Address	54-39-68-1b-27-4d	
Device Type	EdgeMarc 304	
You will need to ena to your manufacture	ble TLS1.2 for SIP registration and enable SRTP for secure calling on your IP phone. Please refer r's instructions for these processes.	
You'll need following inf Supported Devices to vi	ormation for manual provisioning. For Algo/CyberData Paging/Intercom devices, see Zoom Phone ew the configuration guide.	
ankit shukla		•
SIP Account 1:		
1. SIP Domain: 1000	0587.zoom.us	
2. Outbound Proxy: g	gosip0h.sc.zoom.us:5091	
3. User Name: 27894	1760291355532685	
4. Authorization ID: 2	258417410964	
5. Password: VVCaG	LZr	•
Please download your IP phone if t	DigiCert Global Root CA, DigiCert Global Root G2, DigiCert Global Root G3 hey are not in the trust list of the device.	3 and import to
Note: Please no are provisioned phones. It may	ote that Zoom support team will not be able to troubleshoot or configure IP p I in this manner. Some Zoom Phone features may not work on manually prov vary depending on your desk phone model.	phones that isioned
		Close

TLS Configuration between Ribbon EdgeMarc and Zoom

As mentioned in the Zoom Desk Phone Provisioning page:

Please download DigiCert Global Root CA, DigiCert Global Root G2, DigiCert Global Root G3 and import to your IP phone if they are not in the trust list of the device.

Upload the following certificates to the Ribbon EdgeMarc as follows:

Figure 14: Certificates

Certificate Type:	SSL 🗸
Select Certificate File:	Choose File DigiCertGlo otCA.crt.pem
Select Key File:	Choose File No file chosen
Password:	

The uploaded certificates are as follows:

Figure 15: Certificates added

			Certi	ficat	es		
		Name	Туре	C SR	Password	Certificate	Key
111111	8	Digi_GR	CA Certificate			Download	
111111	8	Digi_GR_G2	CA Certificate			Download	
111111	8	Digi_GR_G3	CA Certificate			Download	

Configuring Supplementary Services Configuration on Zoom

Zoom supports multiple supplementary services. To configure different supplementary services in Zoom, refer to the following links:

- Auto Receptionist: https://support.zoom.us/hc/en-us/articles/360001297663-Getting-started-with-Zoom-Phone-admin-#h_a625f531-94c6-4291-909e-3d68ad685b68
- Call Flip: https://support.zoom.us/hc/en-us/articles/360034613311-Using-Call-Flip
- Shared Line Appearance (SLA) or Call Delegation: https://support.zoom.us/hc/en-us/articles/360032881731
- Shared Line Group (SLG): https://support.zoom.us/hc/en-us/articles/360038850792/

Supplementary Services and Features Coverage

The following checklist depicts the set of services/features covered through the configuration defined in this Interop Guide.

Sr. No.	Supplementary Features/Services	Coverage
1	Basic Registration over TLS	\checkmark
2	Basic Call Setup	\checkmark
3	Basic Call Termination	✓

4	Auto Receptionist (Auto Attendant)	\checkmark
5	Call Hold/Resume	\checkmark
6	Call Transfer - Blind (Cold transfer)	\checkmark
7	Call Transfer - Consult (Warm transfer)	\checkmark
8	Conference	\checkmark
9	Call Waiting	\checkmark
10	Call Queue	\checkmark
11	Shared Line Group (SLG)	\checkmark
12	Shared Line Appearance (SLA) or Call Delegation	\checkmark
13	Call Recording	\checkmark
14	Call Flip	\checkmark
15	Call Park	✓

Legend

\checkmark	Supported
×	Not Supported
N/A	Not Applicable

Caveats

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The following issues were observed during recent interop testing:

- The Message Waiting Indicator (MWI) feature was not available for the Zoom Users behind EM 304 ATA.
 - The EdgeMarc 300 was not responding to the INVITE received from Zoom, as it was receiving this INVITE in fragments.
 - This issue is only observed if the INVITE message is large.
 - The issue is resolved in the EM V16.3 release.
- Noise may occur in a conference scenario, when a PSTN (user behind the EM 300) is involved. The fix will be available in upcoming EM releases.

Support

For any support related queries about this guide, contact your local Ribbon representative, or use the details below:

- Sales and Support: 1-833-742-2661
- Other Queries: 1-877-412-8867
- · Website: https://ribboncommunications.com/about-us

References

For detailed information about Ribbon products & solutions, go to :

https://ribboncommunications.com/products

For information about Zoom products & solutions, go to:

https://zoom.us/

Conclusion

This Interoperability Guide describes successful configuration covering Zoom interop with Ribbon EdgeMarc ATA.

All features and capabilities tested are detailed within this document - any limitations, notes or observations are also recorded in order to provide the reader with an accurate understanding of what has been covered, and what has not.

Configuration guidance is provided to enable the reader to replicate the same base setup - there maybe additional configuration changes required to suit the exact deployment environment.

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