New Rock Technologies, Inc.

MX Voice-Fax Gateway Series

High Availability Configuration Guide

HX4E MX8A MX60 MX120

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Overview

1.1 Function Definition

In the deployment of VoIP network, New Rock MX-Series VoIP Gateway (referred as gateway below) supports **high availability** architecture with **Primary-Standby**, **Active-Standby** and **Load balancing** mode.

Primary-Standby mode

In this mode, a backup SIP proxy server (referred as SIP server) is configured. The gateway will failover to the backup server automatically when the primary server faults. The gateway detects the failure condition of primary server by sending OPTIONS request to it. If the gateway does not receive the response to OPTIONS request, it will failover to the backup server.

After failover to the backup server, the gateway will still send OPTIONS to the primary server all the time. It switches back to the primary server once the response to the OPTIONS request is received.

Active-Standby mode

In this mode, one SIP proxy server (referred as SIP server) functions as the primary server while other SIP servers function as standby servers.

Either of the following conditions could trigger the failover operation of the gateway:

- Not receiving response to the OPTIONS message from the current SIP server to which the gateway sends or receives call traffic;
- Not receiving response to the REGISTER/INVITE message from the current SIP server to which the gateway send or receives call traffic

The administrator can manually switchover the gateway from the current SIP server to the next available one.

The gateway will redirect call traffic to the designated proxy server in responding to the re-INVITE from the server.

Load balancing mode

In this mode, the clustered SIP servers are all working in active status. Under the coarse grained scheme all endpoints of a gateway are allowed to register on one of the designated servers and under the fine grained scheme the endpoints of a gateway are allowed to register on multiple servers, according to the administrator's load balancing plan. The following features are supported with load balancing:

- The gateway as a whole or endpoints search for the designated sever in the server cluster (a list of servers) using REGISTER/INVITE message in forward circular scheme.
- Server failure detetion is supported by gateway sending OPTIONS to each servers, on which the gateway or endpoints are registered on.

• Upon the condition of no response to OPTIONS or REGISTER/INVITE, the gateway will search for the next available servers for the gateway or endpoints and move the calls to them accordingly

The gateway will redirect call traffic to the designated proxy server in responding to the re-INVITE from the server.

1.2 Server Cluster

The server cluster includes one primary SIP proxy server and one backup SIP proxy server under primary-standby mode, one primary SIP proxy server and up to five standby proxy servers under active-standby mode or six active servers under load balancing mode. The address of the SIP server can be configured manually by the administrator or obtained through DNS SRV record. Topology is shown as Figure 1-1.

Figure 1-1 Server cluster



2 Configuration

2.1 Configuring Primary-Standby Mode

Step1	Click Basic > SIP .
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- Step2 Choose the High availability mode to Primary-Standby.
- Step3 Fill primary SIP server IP address or domain name in Registrar server.
- Step4
 Fill backup SIP server IP address in Backup SIP proxy.

 Note: This step is required if an IP address is filled in step 3.

When a domain name is filled in step 3, you can also fill in a backup IP address. This allows the device to failover to the backup IP address if the domain name resolution service fails.

Step5 Enable Primary server heartbeat detection, and configure the OPTIONS request period.

Step6 Click Save.

Figure 2-1 Primary-Standby configuration page

Basic	Lin	Line Trunk		Routing)	Advanced	Call Status	Call Status Logs				
Status	Network	VLAN	System	<u>SIP</u>	MGCP	FoIP						
	Lo	ocal signa	ling port		5060			(Range: 1 - 9999, Default: 5060)				
	In	ncrements	of port nur	nber	5			0				
	R	Registrar server]				
	Proxy server					st:5060		e.g. 168.33.134.51:5000 or www.sipproxy.com:5000				
	Subdomain name											
	Registrar mode				Per line	e	•					
	U	User name										
	Registrar password											
	R	egistratio	n expiration	1	600			s				
igh availability	y											
	M	1ode			Primar	y-Standb	ру 🔻	•				
	Ba	ackup SIP	proxy									
	Pr	rimary ser etection	ver heartbe	at								
	0	PTIONS r	equest peri	od	60			s (Range: 2 - 655	35)			
								Save				

Step7 Click **Basic** > **Network**, make sure DNS server can be obtained automatically or has been specified manually.

Note: This step is required if a domain name is filled in step 3.

Basic Line	Trunk	Routing	Advanced	Call Status	Logs T	ools
Status <u>Network</u>	/LAN System SIP	MGCP FoIP				
	Setup		Obtain an IP address	automati 🔻		
	IP address		192.168.10.109			
	Subnet mask		255.255.255.0			
	Default gateway		192.168.10.1			
	DNS server		Obtained automatic	cally	Specified m	nanually
STUN		-				
	STUN		⊖ Enable ● Di	sable		
			s	ave		

Figure 2-2 DNS server configuration page

2.2 Configuring Active-Standby Mode

Step1 Click Basic > SIP.

- **Step2** Choose the High availability mode to **Active-standby**.
- Step3 Fill primary SIP server IP address or domain name in Registrar server.
- **Step4** Click Add and fill the IP address for the standby SIP proxy server in **Standby SIP server 1**. Repeat it for multiple SIP proxy servers.

Note: This step is required if an IP address is filled in step 3.

When a domain name is filled in step 3, you can also fill in a backup IP address in **Standby SIP server 1**. This allows the device to failover to the backup IP address if the domain name resolution service fails.

- Step5 Select to enable or disable OPTIONS Keep-alive.
 - When OPTIONS Keep-alive is enabled, the following timers need to be configured:

OPTIONS request period: the interval between receiving the response (200) from the SIP server to the previous OPTIONS and sending the next OPTIONS.

OPTIONS request timeout: the period since the sending of the last OPTIONS with no response by the SIP server.

- When OPTIONS Keep-alive is disabled, the gateway will failover to the standby SIP server if there is no response to the REGISTER or INVITE.
- Step6 Click Save.

Basic	Lin	e	Trunk		Routing		Advanced	Call Status	Logs	Tools
Status	Network	VLAN	System	<u>SIP</u>	MGCP	oIP				
	1.		l'an an an t		FOGO			(Rearry 1 0000 Det	(
	LC	ical signa	ing port		5060			(Kange: 1 - 9999, De	raut: 5060)	
	In	crements	of port nur	nber	5		•	0		
	Re	egistrar se	erver							
	Pr	oxy serve	r		localhost	5060		e.g. 168.33.134.51:50	00 or www.sippro	oxy.com:5000
	Su	ıbdomain	name							
	Re	egistrar m	ode		Per line		•			
	Us	ser name								
	Re	egistrar p	assword							
	Re	edistration	n expiration		600			s		
ah availabili	ty	-								
	·				1					
	м	ode			Active-St	andby	•			
	SI	P proxy s	ever setting		Add					
	St	andby SIF	proxy serv	er1				e.g. 168.33.134.53:50	00	
	O	PTIONS K	eep-alive		Enable		Disable			
	0	PTIONS re	equest peri	bd	60			s (Range: 2 - 65535)		
	0	PTIONS re	equest time	out	2000			ms (Range: 1000 - 32	2000), if the respo	onse to OPTIONS is timed o
			equest and	out	switch to t	he sta	indby server.			
	Ac	tive SIP s	erver		127.0.0.1:	5060				
					Switchov	er Sw	itchover manually	to the next available s	server.	
								Save		
								- Save		

Figure 2-3 Active-Standby configuration page



The **Switchover** button provides a means to manually switchover the call traffic from the current SIP server to the next available SIP server.

Step7 Click **Basic** > **Network**, make sure DNS server can be obtained automatically or has been specified manually. See Figure 2-2.

Note: This step is required if a domain name is filled in step 3.

Step8 Click Trunk > Advanced, make sure PSTN failover is disabled.

Note: This step is required if OPTIONS Keep-alive is disabled in step 5.

Basic	Line	Trunk	C	Routing	Advanced	Call Status	Logs	Tools	
	Phone number	Freature	Batch	<u>Advanced</u>					
		Gain to I	IP			() dB		
		Gain to I	PSTN				-3.0 dB		
		Impedar	nce		Complex	0 600 Ω	900 Ω		
		Outpulsi	ng delay		1000	m	s (Range: 100 - 3	(000)	
		Caller ID	detectio	on	Before ringing	g v			
		Ring rela	ау		FXS ring syr	nc with FXO ® FX	(S ring independ	ently	
		Busy line	e handle		O Voice prom	pt 💿 Hand up			
		PSTN fai	ilover						
		Inbound	l first dig	it timeout	24	s	s (Range: 10 - 60, Default: 24)		
		Answer	delay		12	s	s (Range: 10 - 60, Default: 12)		
		Off-hool	k for reje	ection	1000	m	ms (Range: 500 - 5000, Default: 1000)		
		On-hool	k protect	ion time	400 ms (Range: 100 - 5000, Default:				
		Polarity	detection	ı	۲				
		Caller nu	umber se	nding mode	DISPLAY	FROM			
usy detection									
		Busy tor	ne count		3	CJ	ycle (Range: 2 - 5)	
		Tone-on	duratio	n	350	m	s (Range: 30 - 10	000)	
		Tone-of	f duratio	n	350	m	s (Range: 30 - 20)00)	
		Detect d	lual-freq	uency busy tone					
						Save			
						Save			

Figure 2-4 Page to disable PSTN failover

2.3 Configuring Load Balancing Mode

- Step1 Click Basic > SIP.
- **Step2** Choose the High availability mode to **Load balancing**.
- Step3 Fill primary SIP server IP address or domain name in Registrar server.
- **Step4** Click **Add** and fill the IP address for the standby SIP proxy server in **SIP server 1**. Repeat it for multiple SIP proxy servers.

Note: This step is required if an IP address is filled in step 3.

When a domain name is filled in step 3, you can also fill in a backup IP address in **SIP server 1**. This allows the device to failover to the backup IP address if the domain name resolution service fails.

- Step5 Configure OPTIONS Settings and REGISTER Settings.
 - **OPTIONS request period**: the interval between receiving the response (200) from the SIP server to the previous OPTIONS and sending the next OPTIONS.
 - **OPTIONS request timeout**: the period since the sending of the last OPTIONS with no response by the SIP server.

• **REGISTER request timeout**: the period from the sending of the first REGISTER with no response by the previous SIP server to the sending of REGISTER to the next SIP server.

Step6 Click Save.

Figure 2-5 Load balancing configuration page

Basic	Line	Trunk		Routing		Advanced	Call Status	Logs	Tools		
Status	Network VLAN	System	<u>SIP</u>	MGCP	FoIP						
	Local sign	naling port		5060			(Range: 1 - 9999, De	fault: 5060)			
	Incremen	ts of port nu	mber	5		,	0				
	Registrar	server									
	Proxy ser	ver		localho	st:5060		e.g. 168.33.134.51:50	00 or www.sippro	oxy.com:5000		
	Subdomain name										
	Registrar	mode		Per line	e	,	•				
	User name										
	Registrar	password									
	Registrati	on expiratio	n	600			s				
High availabilit	у										
	Mode			Load b	alancing						
	SIP proxy	sever setting	9	Ado	1						
	SIP serve	r1					e.g. 168.33.134.53:50	000			
	OPTIONS	request per	iod	60			s (Range: 2 - 65535)				
	OPTIONS	request time	eout	2000			ms (Range: 1000 - 3	2000), if the respo	onse to OPTIONS is timed out,		
				switch to	o the star	ndby server.	ms (Range: 2000 - 3)	2000), if the respo	onse to REGISTER is timed out.		
	REGISTER	(request tim	eout	switch to	o the sta	ndby server.	(,,			
							Save				

Step7 Click **Basic** > **Network**, make sure DNS server can be obtained automatically or has been specified manually. See Figure 2-2.

Note: This step is required if a domain name is filled in step 3.