

**New Rock Technologies, Inc.**

## **MX Voice-Fax Gateway Series**

# **High Availability Configuration Guide**

HX4

MX8

MX60

MX120

<http://www.newrocktech.com>

Tel: +86 21-61202700

Fax: +86 21-61202704

Document Version: TB0-E002-P



## **Amendment Records**

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**Document Rev. 01** (Jan, 2014 )

**Document Rev. 02** (Mar, 2014 )

Add instruction of load balancing feature

**Document Rev. 03** (Mar, 2014 )

Add instruction of REGISTER timeout configuration

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# 1 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 **active-standby** mode and **load balancing** mode.

### 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; or
- 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 switch over 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:

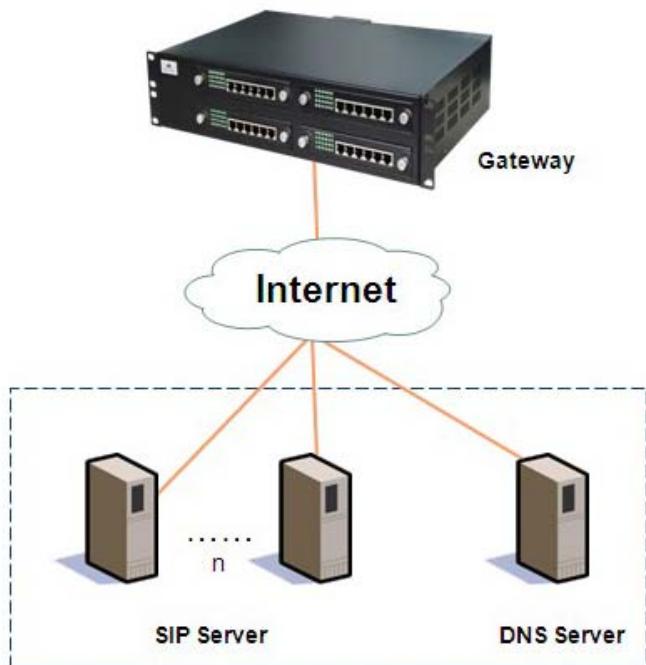
- The gateway as a whole or endpoints search for the designated sever in the server cluster (a list of servers) using REGISTER/INVITE message in forward circular scheme.
- Server failure detection is supported by gateway sending OPTIONS to each servers, on which the gateway or endpoints are registered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

**Figure 1-1 Server cluster**



## 2 Configuring Active-Standby Mode

### 2.1 Enable Active-Standby Feature

Enter the SIP trunk setting page, and click **Basic > SIP > Primary-Standby configuration** and choose **Active-standby**, then submit.

**Figure 2-1 Active-Standby configuration page**

The screenshot shows the 'SIP' configuration page with several tabs at the top: Basic (circled in red), Routing, Line, Advanced, Status, Logs, Tools, and Info. Below the tabs is a navigation bar with Network, System, SIP (circled in red), MGCP, FoIP, and Logout. The main content area contains configuration fields for SIP settings like Signaling port, Auto SIP port selection, Registrar server, Proxy server, User agent domain name, Registration mode, User name, Password, and Registration period. At the bottom, there is a 'Primary-Standby configuration' section with a dropdown menu set to 'Active-Standby' (circled in red). A 'Submit' button is located at the bottom right of this section.

### 2.2 Set Standby SIP Servers

The gateway supports two ways to obtain standby SIP server address:

- IP address
- Domain name

#### 2.2.1 Configuring the IP Address of SIP Servers

Note: the IP address of the primary SIP server is configured on the top half of the SIP page.

Here are the steps to configure the IP addresses of the standby SIP servers:

- Step1** Ensure that active-standby feature is enabled.
- Step2** Fill primary SIP server IP address in **Registrar server**, and then submit.
- Step3** Click **Add** and fill the IP addresses for the standby SIP servers in **Standby SIP servers**.

**Figure 2-2 Page to set registrar server**

The screenshot shows the 'Basic' tab selected in the top navigation bar. The main configuration area is titled 'Network'. Under 'Registrar server', the IP address '192.168.11.4' is listed and circled in red. Other fields include 'Signaling port' (5060), 'Auto SIP port selection' (Off), 'Proxy server' (192.168.11.4), 'User agent domain name', 'Registration mode' (Per line), 'User name', 'Password', and 'Registration period' (600). Below this is a section for 'Primary-Standby configuration' with tabs for 'High availability' (Active-Standby) and 'SIP server cluster(standby)'. Under 'SIP proxy sever setting', there is an 'Add' button circled in red. A table lists 'Standby SIP proxy server 1' (192.168.11.8:5060) and 'Standby SIP proxy server 2' (192.168.11.106:5060). Further down are sections for 'Failover' (Fault condition: No response to OPTIONS request), 'OPTIONS request period' (2), 'OPTIONS request timeout' (1000), and 'Switchover' (Active SIP server: 192.168.11.8:5060). A 'Submit' button is at the bottom.

## 2.2.2 Configuring the Domain Name of the Primary Server

In case the primary SIP server is located through the domain name, the steps below should be follows:

- Step1** Ensure that active-standby feature is enabled.
- Step2** Fill registrar server domain name in **Registrar server**, then submit.
- Step3** Click **Basic > Network**, check **Enable** in **DNS**, fill IP address in **Primary server**, and then submit.

**Figure 2-3 Page to set DNS server**

The screenshot shows the 'Basic' tab selected in the top navigation bar. The main configuration area is titled 'Network'. Under 'DNS', the 'Enable' checkbox is checked and circled in red. The 'Primary server' is set to '192.168.2.5'. There is also a 'Secondary server' field. Below this is a section for 'SNTP' with fields for 'Primary server' (198.60.22.240), 'Secondary server' (133.100.9.2), and 'Time zone' ((GMT+08:00) Beijing). A 'Submit' button is at the bottom.

## 2.3 Set the Failover Condition

You should choose one of the following conditions:

- No response to OPTIONS message
- No response to REGISTER/INVITE message

### 2.3.1 No Response to OPTIONS

When this condition is chosen 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.

**Figure 2-4 Page to set failover condition**

The screenshot shows the 'Primary-Standby configuration' page. Under 'SIP server cluster(standby)', two servers are listed: 'Standby SIP proxy server 1' (IP: 192.168.11.8:5060) and 'Standby SIP proxy server 2' (IP: 192.168.11.106:5060). In the 'Failover' section, the 'Fault condition' is set to 'No response to OPTIONS request'. Two configuration fields are highlighted with red circles: 'OPTIONS request period' (set to 2) and 'OPTIONS request timeout' (set to 1000). The 'Switchover' section shows the 'Active SIP server' as 192.168.11.8:5060. A 'Switchover' button is present. At the bottom is a 'Submit' button.

### 2.3.2 No Response to REGISTER/INVITE

When this condition is chosen, the gateway will failover to the standby SIP server if there is no response to the REGISTER or INVITE.



**Note** When high availability feature is enabled, the PSTN failover feature in **Advanced > Trunk** should be disabled as shown in Figure 2-6.

**Figure 2-5 Page to disable PSTN failover**

Basic	Routing	Line	Advanced	Status	Logs	Tools	Info																										
<a href="#">System</a>   <a href="#">Security</a>   <a href="#">White list</a>   <a href="#">Media stream</a>   <a href="#">SIP</a>   <a href="#">Line</a>   <a href="#">Trunk</a>   <a href="#">RADIUS</a>   <a href="#">Encryption</a>   <a href="#">Tones</a>   <a href="#">Feature codes</a>   <a href="#">Logout</a>																																	
<table border="1"> <tr> <td>Gain to IP</td> <td>0(dB) <input type="button" value="▼"/></td> </tr> <tr> <td>Gain to PSTN</td> <td>-3(dB) <input type="button" value="▼"/></td> </tr> <tr> <td>Impedance</td> <td>Complex <input type="button" value="▼"/></td> </tr> <tr> <td>Outplusing delay</td> <td>600 <small>0-20000(ms), default 400</small></td> </tr> <tr> <td>Ring relay</td> <td><input type="radio"/> FXS ring sync with FXO <input checked="" type="radio"/> FXS ring independently</td> </tr> <tr> <td>Busy line handle</td> <td><input type="radio"/> Voice prompt <input checked="" type="radio"/> Hand up</td> </tr> <tr> <td>PSTN failover</td> <td><input checked="" type="checkbox"/></td> </tr> <tr> <td>Caller ID detection mode</td> <td>After ringing A <input type="button" value="▼"/></td> </tr> <tr> <td>Inbound first digit timeout</td> <td>24 <small>10-60(s), default 24. Timeout of collecting DTMF on FXO for inbound call</small></td> </tr> <tr> <td>Answer delay</td> <td>12 <small>10-60(s), default 12. Also see " Connect signal delay " in page " Line &gt; Trunk "</small></td> </tr> <tr> <td>Off-hook for rejection</td> <td>1000 <small>500-5000(ms), default 600</small></td> </tr> <tr> <td>On-hook protection time</td> <td>400 <small>100-5000(ms), default 400</small></td> </tr> <tr> <td>Polarity detection</td> <td><input checked="" type="checkbox"/></td> </tr> </table>								Gain to IP	0(dB) <input type="button" value="▼"/>	Gain to PSTN	-3(dB) <input type="button" value="▼"/>	Impedance	Complex <input type="button" value="▼"/>	Outplusing delay	600 <small>0-20000(ms), default 400</small>	Ring relay	<input type="radio"/> FXS ring sync with FXO <input checked="" type="radio"/> FXS ring independently	Busy line handle	<input type="radio"/> Voice prompt <input checked="" type="radio"/> Hand up	PSTN failover	<input checked="" type="checkbox"/>	Caller ID detection mode	After ringing A <input type="button" value="▼"/>	Inbound first digit timeout	24 <small>10-60(s), default 24. Timeout of collecting DTMF on FXO for inbound call</small>	Answer delay	12 <small>10-60(s), default 12. Also see " Connect signal delay " in page " Line &gt; Trunk "</small>	Off-hook for rejection	1000 <small>500-5000(ms), default 600</small>	On-hook protection time	400 <small>100-5000(ms), default 400</small>	Polarity detection	<input checked="" type="checkbox"/>
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## 2.4 How to Manually Perform Switchover

On the Wen GUI of the gateway, the **Switchover** button provides a means to manually switchover the call traffic from the current SIP server to the next available SIP server.

# 3 Configuring Load Balancing Mode

## 3.1 Enable Load Balancing Feature

Enter the SIP trunk setting page, and click **Basic > SIP > Primary-Standby configuration** and choose **Load balancing**, then submit.

**Figure 3-1 Load balancing configuration page**

Signal port	5060	1-9999,default 5060
Auto SIP port selection	Off	1-10:Local SIP port will auto select, based 5060 increasing the value
Registrar server	10.128.3.90:8989	
Proxy server	10.128.3.90:8989 e.g. 168.33.134.50:5060 or www.sip.com:5060	
User agent domain name	e.g. www.gatewaysip.com	
Registration mode	Per line	▼
User name		
Password	You may obtain it from service provider	
Registration period	15	15-86400(s), default 600

[Primary-Standby configuration](#)

High availability  
  Off  
  Active-Standby  
  Load balancing

[Submit](#)

## 3.2 Set SIP Servers

Refer to [2.2 Set Standby SIP Servers](#).

## 3.3 Configure OPTIONS Settings

In the active balancing mode, 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.

**Figure 3-2 Page to configure OPTIONS settings**

Primary-Standby configuration

High availability	Load balancing <input checked="" type="checkbox"/>
SIP server cluster(standby)	
SIP proxy sever setting	<input type="button" value="Add"/>
SIP proxy server1	192.168.11.8:5060 e.g. 168.33.134.53:5060
SIP proxy server2	192.168.11.106:5060 e.g. 168.33.134.53:5060
OPTIONS setting	
OPTIONS request period	2 s(rang:1-86400)
OPTIONS request timeout	1000 ms(rang:1000-32000), if the response to OPTIONS is timed out, switch to the standby server.
REGISTER setting	
REGISTER request timeout	2000 ms(rang:2000-32000), if the response to REGISTER is timed out, switch to the standby server.
Active server list	
1	192.168.11.8:5060
<input type="button" value="Submit"/>	

### 3.4 Configure REGISTER Settings

In the active balancing mode, the following time need to be configured:

- **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.

**Figure 3-3 Page to configure REGISTER settings**

Primary-Standby configuration

High availability	Load balancing <input checked="" type="checkbox"/>
SIP server cluster(standby)	
SIP proxy sever setting	<input type="button" value="Add"/>
SIP proxy server1	192.168.11.8:5060 e.g. 168.33.134.53:5060
SIP proxy server2	192.168.11.106:5060 e.g. 168.33.134.53:5060
OPTIONS setting	
OPTIONS request period	2 s(rang:1-86400)
OPTIONS request timeout	1000 ms(rang:1000-32000), if the response to OPTIONS is timed out, switch to the standby server.
REGISTER setting	
REGISTER request timeout	2000 ms(rang:2000-32000), if the response to REGISTER is timed out, switch to the standby server.
Active server list	
1	192.168.11.8:5060
<input type="button" value="Submit"/>	

### 3.5 Active Server List

All the SIP servers, on which the gateway or endpoints are registered on, will be listed in active server list.