

Grandstream Networks, Inc.

GRP26XX Carrier-Grade IP Phones

Server Redundancy Guide





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SUPPORTED DEVICES

Model	Supported	Firmware	
Carrier-Grade IP Phones GRP26XX Series			
GRP2612/GRP2612P/ GRP2612W			
GRP2613	Yes	1.0.0.31 or higher	
GRP2614			





INTRODUCTION

A redundant server is mostly used to assure the reliability of an end point's service when it loses connectivity with the primary server.

Configuring a redundant server is recommended for medium and large VoIP deployment installations.

Users can then keep using the service on their end points when the main server cannot be reached, service is down or when administrators need to do maintenance tasks on it.

This guide will outline the use and configuration of redundant SIP server on Grandstream GRP26XX.

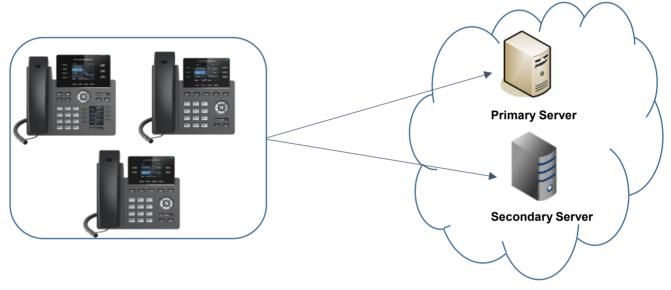


Figure 1: Server Redundancy Diagram





SIP SERVER REDUNDANCY

The GRP26XX will send REGISTER requests and SUBSCRIBE messages (except for message waiting) to both primary and secondary SIP servers for the same account, when both primary and secondary SIP servers are configured.

When making a call, the phone will use the registered primary SIP server first. If not available, the secondary SIP server will be used instead.

Requirements

- A SIP end point supporting primary and secondary SIP server configuration options under its SIP account(s)/Profile(s).
- Two SIP servers having the same extension's credentials.

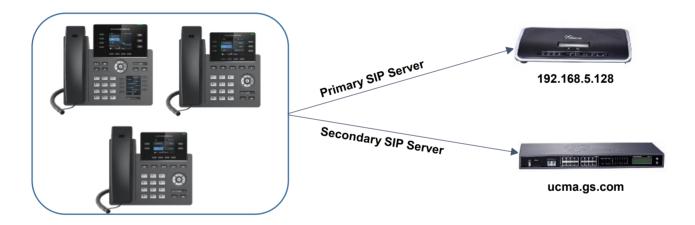


Figure 2: SIP Server Redundancy Diagram

Configuration on GRP26XX

The following shows usage and configuration of primary and secondary SIP server from he Web GUI:

- Go to Accounts → Account X → General Settings.
- 2. Enter the "IP Address: Port" or "FQDN: Port" of your primary SIP server in SIP Server field.
- Enter the "IP Address: Port" or "FQDN: Port" of your secondary SIP server in Secondary SIP
 Server field.





Note: Do not configure same SIP Server address in Primary and Secondary SIP Servers fields.



Figure 3: Account General Settings

4. (Optional) Administrator can also change Register Expiration and Reregister before Expiration values so the end point can check and refresh its registration accordingly with set values (in minutes for Register Expirations and in seconds for Reregister before Expiration).
In the below figure Register Expiration is set to 60 minutes, while Reregister before Expiration will not be used (set to 0 second).

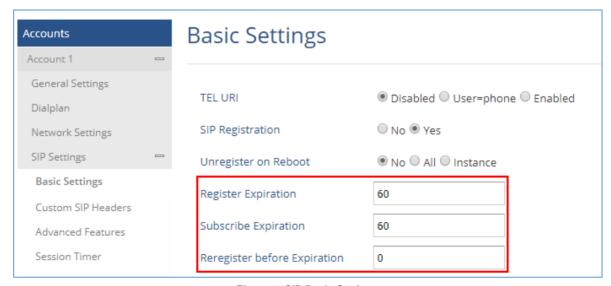


Figure 4: SIP Basic Settings





Phone Behavior against Servers Availability

In above example, GRP26XX will send two SIP REGISTER requests to the IP/FQDN configured in **SIP** Server and Secondary SIP server fields.

Case1: Both Servers Reachable

If both SIP Server and Secondary SIP Server are reachable, the phone will register on both servers.

The phone will always use the primary server for calls and refresh its registration each *Register Expiration* period (60 minutes in above example) to ensure that both servers are still reachable.

Case 2: Primary Server Not Responsive

If primary **SIP Server** is not responsive, the phone will use **Secondary SIP Server** for phone services instead (including making/receiving calls).

Flow Examples

The following figure shows SIP flow example between Grandstream IP phone (GRP2613 in this example) and primary/secondary SIP Servers. The flow shows successful registration on both primary and secondary SIP servers (case 1: both servers reachable) also when the primary SIP server becomes unresponsive (case 2: primary SIP server not responsive) to the SIP INVITE.





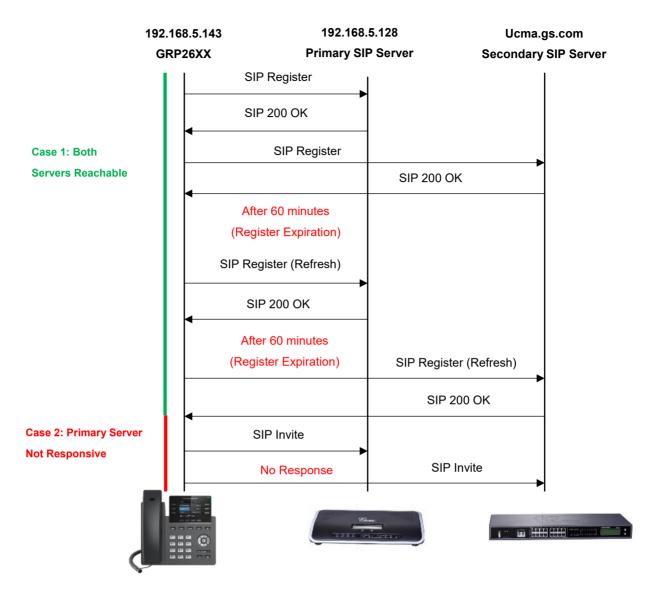


Figure 5: SIP Registration and Invite with Primary and Secondary SIP Servers Flow

Note: We assume in above scenarios that SIP servers are not challenging the SIP register with 401 or 407.

