OpenVox



GSM Gateway Connect with Elastix® Server QUICKSTART GUIDE

This document applies to OpenVox GSM Gateway VS-GW1200-4G/VS-GW1202-8G and VS-GW1600 series. There are two RJ45 Network ports, ETH1 and ETH2. If you choose ETH1, you can access Board 1 only, and access other boards with the same IP address, different port numbers. This will help to avoid IP conflict. If you choose ETH2, you can access different Boards with different IP addresses.

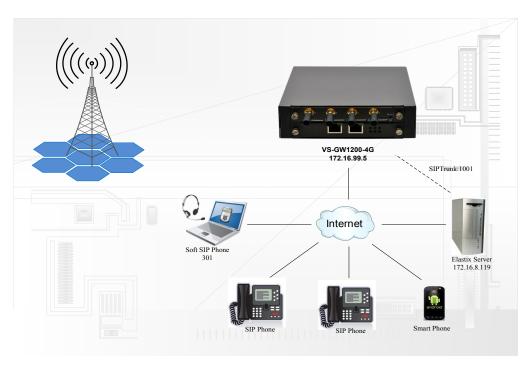
VoxStack provides 2 working modes: Stand-alone and Cluster.

⇒ Stand-alone: A single IP address manages one GSM modules (4 ports).

Stack Num	IP	Username	Password
1	172.16.99.1	admin	admin
2	172.16.99.2	admin	admin
3	172.16.99.3	admin	admin
4	172.16.99.4	admin	admin
5	172.16.99.5	admin	admin

⇒ Cluster: A single IP address manages up to 5 GSM modules (up to 20 ports).

Default IP: 172.16.99.1 User Name: admin Password: admin



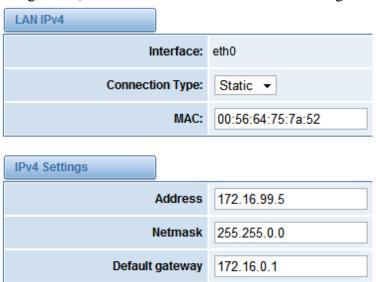
www.openvox.cn 1/5





Step 1. Set Network Parameters in Web

If your system topology like the figure described, please enter the gateway default IP address to login web, and click "NETWORK—>LAN Settings" to set network parameters such as IP.

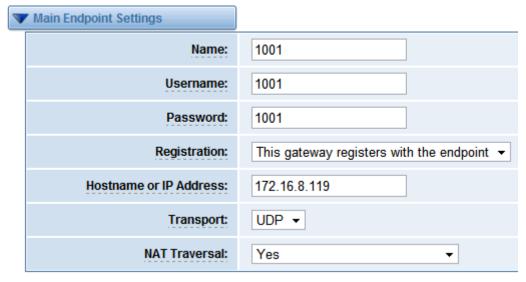


Save your changes. Please type in your DNS server in "DNS Server Address".



Step 2. Create a SIP Endpoint in Web

Please select "SIP—>SIP Endpoints—>Add New SIP Endpoint" to set SIP trunk. The following figure shows detail information about how to set it.



About other parameters in SIP, please set by your requirements for there is no need to set them in simple calls.

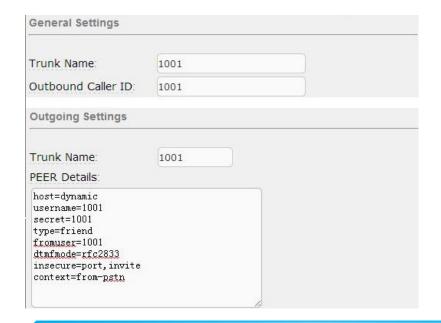


Step3. Create a SIP Trunk in Elastix® Server

Please login your Elastix® server to create a SIP trunk(1001). In Elastix® server web, please choose "PBX—>Trunks—>Add SIP Trunk" to set like that:

www.openvox.cn 2/5

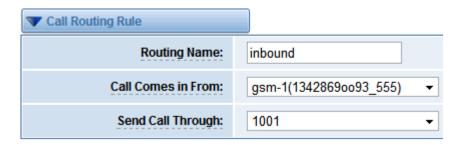






Step 4. Set Routing Rules in Web

Click "ROUTING—> Call Routing Rules—> New Call Routing Rule" to set outbound and inbound routing rules like the following:



Save the inbound call routing rules, please set the outbound rules as introduced. In order to make all calls successfully, please enable and set failover function in advanced routing rule like that:



Please save all your changes to make effect.

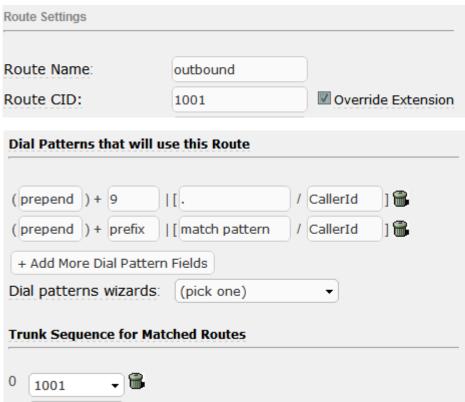
www.openvox.cn 3/5



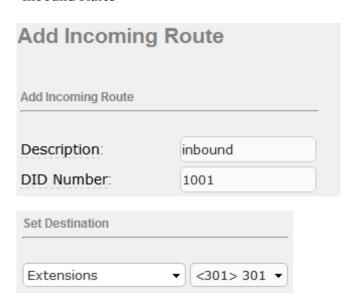


Step 5. Configure Routings in Elastix

Outbound Rules



Inbound Rules



www.openvox.cn 4/5

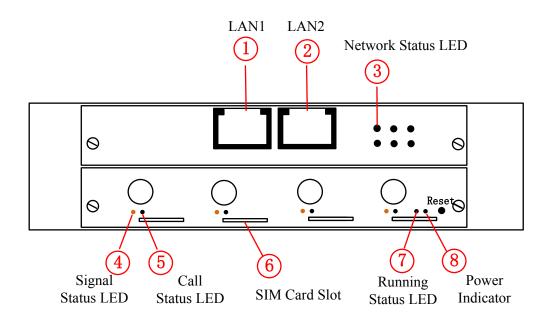




Step 6. Register a SIP extension by software

Taking advantage of SIP software such as Xlite, eyeBeam to register a SIP extension(301).

Front Panel



C 1E1 1		
Green and Flash	Module Initiating	
Red and Flash	No SIM Card	
Red and No-flash	Worst Signal Quality	
Yellow and No-flash	Medium Signal Quality	
Green and No-flash	Best Signal Quality	
Flash	Communicating	
Blind	Normal	
Green and Flash	Work Normally	
Always Green	Supply Power	
dicators flash.		
	Red and No-flash Yellow and No-flash Green and No-flash Flash Blind Green and Flash	

www.openvox.cn 5/5