



OpenVox GSM Gateway Clustering

QUICKSTART GUIDE

This document applies to OpenVox GSM Gateway VS-GW1600 (including 8G/12G/16G/20G). This is an example with 20G. 8G/12G/16G/20G GSM Gateway will be auto-matically assigned to 2/3/4/5 different IPs in each 4 GSM channels. You can access each GW in cor-responding IP from 172.16.99.1 to 172.16.99.5. Each 4 channels are separated, which is able to avoid crashing during the production. It is also available to configure a cluster to use one Master module to control all other Slave modules. Then you can control all 20 gsm ports with a single IP address. Here is a demo to show you how to configure the cluster.

Default ID. 172 16 00 1	Versions	Modules	IP Address
Default IF: 172.10.99.1	VS-GW1200-4G	1 * 4G	172.16.99.1
T I I	VS-GW1600-8G	2 * 4G	172.16.99.1-2
Username: admin	VS-GW1600-12G	3 * 4G	172.16.99.1-3
Password : admin	VS-GW1600-16G	4 * 4G	172.16.99.1-4
	VS-GW1600-20G	5 * 4G	172.16.99.1-5

There are two LAN ports, please connect the gateway to Internet through one of LAN port and make sure connectivity by LED status.

Configuration

 \Rightarrow Configure options in GUI:

- 1. Factory Reset All Modules;
- 2. Cluster Settings;
- 3. Network Parameters;
- 4. SIP Settings;
- 5. Routings;
- 6. Call Test
- \Rightarrow Create a SIP trunk, and configure VoIP gateway in SIP server
- \Rightarrow Register SIP extensions
- \Rightarrow Call Test





Step 1. Factory Reset All Modules

Before setup cluster, you need to upgrade the latest firmware and factory reset all modules. This can be done via web GUI or hardware RST button.

Reset the device via WEB GUI:

Click "SYSTEM—>Tools—>Factory Reset

SYSTEM	penVox Solution
Reboot Tools	
his will turn off your gateway and then turn it back on. This will drop all current calls.	System Reboot
This will restart asterisk. This will drop all current calls.	Asterisk Reboot
Update Firmware lew system file: 选择文件 未选择文件	System Update
Upload Configuration	File Unload
Backup Configuration	The open
Current configuration file version: 1.0.3	Download Backup
Restore Configuration	
	Eactory Pasat

Reset the device via Hardware RST botton:

TEL:+86-755-82535461 FAX:+86-755-83823074





Notice:

Press the RST botton for 8 seconds, then the system will restart and facotry reset the device. About 90 second later, you can login the device with default IP address list above, then we can start to do cluster for managing the device through a single IP address.



Step 2. Cluster settings



Click "SYSTEM—> Cluster—> Set Deafult" to setup cluster for all VS-GWM400G modules:

SYSTEM		Free	e Comm	unlca	tion OpenVox S	olution
anaged Mode						
Mode:	Master		Set Default			
Password:	9999					
Master IP(Local IP):	192.168.9	D				
	Board-2	Original IP:	172.16.99.2	Target IP:	192.168.9.2	
01	Board-3	Original IP:	172.16.99.3	Target IP:	192.168.9.3	
Slaves IP List:	Board-4	Original IP:	172.16.99.4	Target IP:	192.168.9.4	
	Board-5	Original IP:	172.16.99.5	Target IP:	192.168.9.5	
Remain Origianl IP address.	ON					



Mode:

Stand-alone ------ Work individually as a 4 ports GSM gateway; Master ----- Act as a Master of all VS-GWM400G modules, it will control all slave modules. one master, others are slave; Slave -----Act as a Slave, it is under Master's control.

Password:

It is for comunication between Master and Slaves.

Master IP (Local IP):

This IP is for internal communication between Master and Slaves. It is different from external IP for access to Administrator console via WEB GUI. You can setup it whatever you want, just make sure it will not conflict in LAN.

Slave IP List:

There are 2 kinds of IP here, Original IP is for external access to each module, default IP is 172.16.99.1-5, Target IP is for internal communication in clustering.

If Master IP settled, Target IP will be settled automaticly. When clicked "Apply", Target IP will be put into effect for each module based on Original IP. Here cluster done!

Remain Original IP Address:

If 'ON', remain external IP for each module; If 'OFF', only remain external IP for Master, Slave not.





Step 3. Setup Network Parameters

After cluster settings finished, you can change the IP address for Master module, both internal IP and public IP are OK. Please enter the default IP of gateway address in your browser to login Administrator Console, and click "NETWORK—>LAN Settings" to set network parameters such as IP Address. Choose "Static" Type for eth0 Interface, fill in your IP address, then "Save" your changes.

WIRELESS GATEWAY	YSTEM GSM SIP ROUTING METWORK ADVANCED LOG
NETWORK	Free Communication OpenVox Solution
LAN IPv4	
Interface:	eth0
Туре:	Static
MAC:	A0:98:05:01:08:39
IPv4 Settings	
Address:	172.16.99.1
Netmask:	255.255.0.0
Default Gateway:	172.16.0.1
Reserved Access IP	
Reserved Access IP Enabled:	
Reserved Access IP Enabled: Reserved Address:	ON

Save

Reserved Access IP:

This option is reserved to access to gateway if all passwords missed just in case. It is not changeable. You can choose to Enable it or not, if "OFF" selected, the IP will disappear.

you also need to setup DNS server in DNS settings for time synchronization.

(WIRELESS GATEWAY)	C	AN Settings WAN S	ettings ONS Settings D	DNS Settings Toolkit	
NETWORK		Free C	ommun]ca	tion OpenVox S	iolution
DNS Servers					
DNS Servers DNS Ser	rver 1: 8.8.8.8				
DNS Servers DNS Ser DNS Ser	rver 1: 8.8.8.8				
DNS Servers DNS Ser DNS Ser DNS Ser	rver 1: 8.8.8.8 rver 2: .				

Save



Step 4. SIP Settings

Create a SIP Endpoint for connecting VoxStack and SIP server

Please enter the IP of gateway in your browser to login Administrator Console, and click "SIP—>SIP Endpoints—>Add New SIP Endpoint" to build a sip trunk.

	(SIP Endpoints	Advanced SIP Settings	
	Free	<i>Commun</i> [cation	OpenVox Solution
Endpoint Name	Registration	Credentials	OpenVox Solution
Endpoint Name	Registration client	Credentials 1025@172.16.65.65	OpenVox Solution

Add New SIP Endpoint

Here is a sample to show you how to create SIP trunks for connecting VoxStack to a VOS3000 VoIP Operation Platform with Register Mode and UnRegister Mode.

• Register Mode: Register VoxStack to SIP server with username and password

Name:	VOS3000	
Username:	2002 Anonymous	
Password:	2002	
Registration:	This gateway registers with the endpoint	
Hostname or IP Address:	172.16.110.110	
Transport:	UDP 🗶	
NAT Traversal:	Yes	

Save Cancel

• UnRegister Mode: Connect VoxStack to SIP server without authtication, just IP to IP. Add a New SIP Endpoint

Password: Registration: None Hostname or IP Address: 172.16.110.110 Transport: UDP • NAT Traversal: Yas	Password:	
Registration: None Hostname or IP Address: 172.16.110.110 Transport: UDP • NAT Traversal: Yes		
Hostname or IP Address: 172.16.110.110 Transport: UDP x NAT Traversal: Vas	Registration:	None
Transport: UDP 💌	Hostname or IP Address:	172.16.110.110
	Transport:	
	NAT Traversal:	Yes



Notice:

If you need to build multiple sip trunks for connecting VoxStack to a same SIP server,

fromuser options must be enabled in the SIP peer setting when you create SIP trunk on

your SIP server.

Set codec priority for all SIP endpoints

Click "SIP—>Advanced SIP Settings—>Codec Settings" to set codec priority, it is a general setting, will be applied to all SIP endpoints.

VoxStack sys	STEM GSM SP ROUTING NETWORK ADVANCED LOGS
SIP DETAILS	Free Commun Cation OpenVox Solution
Networking Parsing and Compatibility	
Security Modia	
Codec Settings	
Codec Priority 1:	G.729 💌
Codec Priority 2:	G.711 u-law 💌
Codec Priority 3:	G.711 a-law 💌
Codec Priority 4:	GSM
Codec Priority 5:	G.722 ×
Codec Priority 6:	G.726 ×

Save



If you want to use a specific codec, such as G729, then you have to put G.729 in top priority. Otherwise VoxStack will not transfer calls with the codec you expect.



Step5. Setup Routings to handle call traffic

Create a GSM group

VoxStack provides 4/8/12/16/20 gsm ports for GSM network access, with Cluster mode, it is easy to separate all those ports to different GSM groups with different policy to route calls. You can also add sip endpoints to a SIP group. Thus make routings truely flexible and simple.

Click "ROUTING—>Groups—>New Group" to create GSM group or SIP group.



WIRELESS GATE	WAY		Call Routing Rule	s Groups		
ROUTING DETAILS	Å		Free	Communic	ation ope	nVox Solution
Group Name	Type	Policy	Members			Actions

Here is a sample show you how to create a GSM group to contain all 20 ports.

outing Groups		
Group Name:	GSM_ALL	
Туре:	(GSM)	
Policy:	Roundrobin	
Members	NO. ✓ All 1 ✓ Board-1-gsm-1 2 Ø bard-1-gsm-2 3 Ø baard-1-gsm-3 4 Ø baard-2-gsm-3 5 Ø baard-2-gsm-1 6 Ø baard-2-gsm-2 7 Ø baard-2-gsm-3 8 Ø baard-2-gsm-4 9 Ø baard-2-gsm-1 10 Ø baard-3-gsm-1 10 Ø baard-3-gsm-2 11 Ø baard-3-gsm-2 11 Ø baard-4-gsm-3 12 Ø baard-4-gsm-1 13 Ø baard-4-gsm-2 14 Ø baard-4-gsm-2 15 Ø baard-4-gsm-3 16 Ø baard-4-gsm-4 17 Ø baard-4-gsm-4 18 Ø baard-5-gsm-2 19 Ø baard-5-gsm-3	

Save Cancel

Group Name:

It is for identifying different group in case multiple groups exists.

Type:

There are 2 types of group, GSM and SIP.

Policy:

7 kinds of policies are available for chosing GSM or SIP channel, Ascending, Descending, Roundrobin, Revers Roundrobin, Least Recent, Fewest Calls and Ramdon. You can choose the policy based on your needs, normally, Roundrobin and Reverse Roundrobin recommended.

Members:

You can choose any number of gsm ports to the group you want to create.

Click "Save" to apply the changes. You can follow the same step to create SIP groups if needed.



Create Call Routing Rules

VoxStack provides Flexible Routing rules to handle calls, from SIP to GSM, GSM to SIP, SIP to SIP and GSM to GSM, you can achieve whatever you want.

Click "ROUTING—>Call Routing Rules—>New Call Routing Rule" to create Routing rules.

			Call Routing Rules Groups		
RO	UTING				
Move	Order	Rule Name	Free Com	nun (cation .	Actions
Move	Order	Rule Name	From 1025	TO GSM ALL	Actions

• Create a routing rule from VOS3000 to GSM group, calls from VOS3000 will be sent to all gsm ports with Round robin policy.

outing Rule		
Routing Name:	VOS2GSM	
Call Comes in From:	VOS3000	
Send Call Through:	GSM_ALL Y	

prepend) + prefix	I match pattern	/ Callerld	1 💢
---------	------------	-----------------	------------	-----

Now focus on the dial Patterns:

Dial Patt	erns that wi	Il use this Route		
0086) + 88	[01.	/ 123456789] 💥	

Prepend ----- 0086 Prefix ----- 88 Match Pattern ----- 01. CallerId ----- 123456789

Notice:



A dot (.) matches one or more dialed digits.

It means only calls with CallerID '123456789' and CalleeID '8801.' from VOS3000 can be sent to GSM group, after modified by the dial pattern, CalledID will be changed to '008601.', if leave the blank empty, then corresponding parameter will be unlimited.



• Create another routing rule from GSM group to VOS3000, then calls from all gsm ports will be sent to VOS3000 through sip trunk 2002.

Routing Name:	GSM2VOS	
Call Comes in From:	GSM_ALL	
Send Call Through:	VOS3000	



Step6. Call Test

Apply all changes on VOS3000 and GSM gateway, then you can try to make calls. Taking advantage of SIP software such as Xlite/Eyebeam to register a SIP extension 123456789 on VOS3000 server.

• Test call from VOS3000 to GSM geteway

Use Extension 123456789 to call '8801+digits', then you will reach '008601+digits', you can check it on the asterisk log of VoxStack.

• Test call from GSM geteway to VOS3000

Use your mobile to call numbers of SIM cards on GSM gateway, then calls will be routed

to VOS3000, you need to setup a Mapping Gateway for VoxStack to handle the calls.