

iAG800 V2 Series Analog Gateway

Overview

OpenVox iAG800 V2 series Analog Gateway, an upgrade product of the iAG Series, is an open source asterisk-based Analog VoIP Gateway solution for SMBs and SOHOs. With friendly GUI and unique modular design, users may easily setup their customized Gateway. Also secondary development can be completed through AMI (Asterisk Management Interface).

The iAG800 V2 Analog Gateways are comprised of six models: iAG800 V2-4S with 4 FXS ports, iAG800 V2-8S with 8 FXS ports, iAG800 V2-4O with 4 FXO ports, iAG800 V2-8O with 8 FXO ports, iAG800 V2-4S4O with 4 FXS ports and 4 FXO ports, and iAG800 V2-2S2O with 2 FXS ports and 2 FXO ports.

The iAG800 V2 Analog Gateways are developed for interconnecting a wide selection of codecs including G.711A, G.711U, G.729A, G.722, G.726, iLBC. iAG800 V2 series use standard SIP protocol and compatible with Leading VoIP platform, IPPBX and SIP servers. Such as Asterisk, Issabel, 3CX, FreeSWITCH, BroadSoft and VOS VoIP operating platform.





Key features

- 2/4/8 FXS/FXO ports
- Support SNMP and TR069
- Support IPv4 and IPv6
- 3km Line Length(MAX)

- Multiple codecs: G.711A/U,G.723.1,G.729A/B and iLBC
- Support easy management
- Fully Compatible with leading SIP based VoIP Telephony System

Physical Interfaces

Capacity

- 2/4/8 FXS with RJ11
- 2/4/8 FXO with RJ11

Ethernet Interfaces:

- 1 * WAN, 10/100Mbps, RJ-45
- 1 * LAN, 10/100Mbps, RJ-45

Voice & FAX

- G.711A, G.711U, G.729A, G.722, G.726, iLBC
- Comfort Noise Generation(CNG)
- Modem/POS
- Adaptive (Dynamic) Jitter Buffer
- Programmable Gain Control
- DTMF mode: Signal/RFC2833/INBAND
- T.38/Pass-through
- Echo Cancellation(G.168), with up to 128ms

Call Features

- Call Waiting
- Blind Transfer
- Attend Transfer
- Call Forward on Busy
- Call Forward on No Reply
- Unconditional Call Forward
- Warm/Immediately Hotline
- Message Waiting Indicator
- SIP registration to different servers
- Call Hold
- Do-not-disturb
- 3-Way Conference
- Speed Dial

FXS

- Connector: RJ11
- Dial Mode: DTMF and Pulse
- Pulse: 10 and 20 PPS
- Caller ID: DTMF/FSK CLI Presentation
- Max Cable Length: 3 km
- Reversed Polarity Programmable Call Progress Tone

FXO

- Connector: RJ11
- Dial Mode: DTMF/Pulse Dialing
- Caller ID: FSK, DTMF
- Polarity Reversal
- Answer Delay
- Busy Tone Detection
- No Current Detection

Software Features

- Hunting Group
- Web ACL
- Telnet ACL
- Action URL
- PPPoE
- OpenVPN
- Digitmap
- Routing Rules based Prefixes
- Caller/Called Number Manipulation

VoIP

Protocol:

- SIP v2.0 (UDP/TCP), RFC3261
- SDP,RTP(RFC2833), RFC3262, RFC3263, RFC3264, RFC3265, RFC3515, RFC2976, RFC3311
- RTP/RTCP, RFC2198, RFC1889
- IPv4 and IPv6
- Outbound Proxy
- RFC2806 TEL URI
- RFC3581 NAT,rport
- VLAN 802.1P/802.1Q
- RFC4028 Session Timer
- DNS SRV/A Query/NATPR Query
- NAT:STUN, Static/Dynamic NAT

Maintance

- SNMP v1/v2/v3
- Auto Provisioning
- Web Configuration
- Backup/Restore
- Firmware Upgrade via Web
- CDR
- TR069
- IPoE
- Syslog(Emerg,alert, critical,error warning,notice,info, debug)
- Ping/Tracert Test
- Network Capture
- NTP/Daylight Saving Time
- IVR local Maintenance
- Cloud-based Management

OpenVox iAG200/400 Series Analog Gateway

Product Name iAG800-202S iAG800-4S iAG800-40 iAG800-404S iAG800-8S iAG800-80

Interfaces

FXS/FXO Ports 4/4 8 8 2/2

2 * 10/100Mbps ports Ethernet port

General Info

Weight 637g

Dimension 190mm*142mm*35mm

Maximum power 12W

Universal Power Supply 12V 2A

Operation Temperature 0°C ~ 50°C

Operation Humidity Range 10% ~ 90% NON-CONDENSING

Storage Temperature Range -20°C ~ 70°C

Short/Long Haul Ring Load 2 REN, up to 3km on 24AWG line

Certification CE



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