

iAG801 Series Analog Gateway

Overview

OpenVox iAG801 is a multifunctional analog gateway that provides 8 FXS ports for seamless connection between IPPBX, fax machines, analog phones and operators. In addition, it has excellent full concurrent voice/fax processing capabilities, strong performance and high stability, and provides high-quality call services for operators, enterprises, call centers and residential users in residential communities.

The iAG801 Analog Gateway, a cost-effective product of the iAG Series, is an ideal analog VoIP gateway solution for SMBs and SOHOs. With friendly GUI and unique design, users may easily customize and configure their gateway. Also secondary development can be completed through AMI (Asterisk Management Interface).

The iAG801 Analog Gateway is developed for interconnecting a wide selection of codecs including G.711A, G.711U, G.722, G.726, G.729A, iLBC. iAG801 series use standard SIP protocol and compatible with leading VoIP platform, IPPBX and SIP servers, such as Asterisk, Issabel, 3CX, FreeSWITCH, Broad-Soft and VOS VoIP operating platform.





Key features

- 8 FXS 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

8 FXS 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

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

Software Features

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

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 iAG801 Series Analog Gateway

Product Name iAG801

Interfaces

FXS/FXO Ports 8

Ethernet port 2 * 10/100Mbps ports

General Info

Weight 400g

Dimension 170mm*98mm*26mm

Maximum power 12W

Universal Power Supply DC 12V/2A

Operation Temperature $0^{\circ}\text{C} \sim 50^{\circ}\text{C}$

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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