



DAG1000-8FXS Analog VOIP Gateway

Overview

DAG1000 series analog gateway are versatile IP-based voice and fax gateway. For small business, the family office, remote office and branch business provide a low cost, simple operation VoIP solutions. Connected analog phone, fax, PBX with standard voice interface and provide quality voice service. DAG1000 series VoIP access gateway use standard SIP protocol and compatible mainstream IPPBX and softswitch platform.

Product pictures




DAG1000-4S



DAG1000-8S

Key benefits

- Primary and secondary SIP account
- Flexible set port group
- IP trunk
- Flexible routing and manipulation
- Data/ Voice/ Management VLAN
- Voice /FAX/Modem
- SNMP
- Firmware update automatically /configure update



Decade focused

Creates the high-quality goods

Physical Interfaces

- Phone Interface: 1/2/4/8 FXS, RJ-11
- Ethernet Interface:
4FXS: 2* RJ-45 10/100Mbps, RJ-45
8FXS: 4* RJ-45 10/100Mbps, RJ-45

Voice & Fax

- G.711A/U law, G.723.1, G.729A/B
- Comfort Noise Generation(CNG)
- Voice Activity Detection(VAD)
- Echo Cancellation(G.168)
- Adaptive (Dynamic) Jitter Buffer
- Hook flash test
- Programmable Gain Control
- T.38/Pass-through/Modem FAX mode,
Up to 14.4kbps
- DTMF mode: Signal/RFC2833/INBAND

Private service

- Inquire LAN port IP *158#
- Inquire WAN port IP *159#
- Get IP address *150*
- Restart*111#
- Factory reset*166*000000#

Protocol

- SIP V2.0 (RFC 3261,3262,3264)
- SDP (RFC 2327)
- REFER (RFC 3515)
- RTP/RTCP (RFC 1889,1890)
- STUN (RFC 3489)
- ARP/RARP (RFC 826/903)
- SNTP (RFC 2030)
- DHCP/PPPoE
- TFTP/HTTP/HTTPS
- DNS/DNS SRV (RFC 1706/RFC 2782)
- VLAN 802.1P/802.1Q

Supplement Service

- Call waiting
- Call transfer(Blind transfer, Attend Transfer ,
Half blind transfer)
- Quick pick
- Call forwarding
- Hotline
- Call hold
- DND
- 3-way conference (1/2/4FXS Support)
- Voice mail
- Direct IP address call



Customer first

Services more intimately

Call & Route

- Port group
- IP trunk
- Primary and secondary SIP account
- 32 inbound/outbound routing
- Number manipulation
- Dial plan set

Network Features

- NAT (Rout and Bridge)
- Static IP, PPPoE, DHCP
- DHCP server
- MAC address clone
- DMZ, Static routing
- IP, TCP, UDP, TFTP, FTP, TCP, ARP, RARP, Ping, NTP, SNTP, Http, DNS, Trancert
- Static/dynamic ARP

Environment

- Power: input:100-240V,50-60Hz
Output: 4FXS :12VDC,1A
8FXS :12VDC, 2A
- power consumption(w): 4S:12,
8S:20w,16S:48w
- temperature(operator): 0 °C ~ 45 °C
(storage): -20 °C ~80 °C
- Humidity: 10%-90% no condensation

Maintain & Upgrade

- Web based configuration
- Telnet configuration
- Data backup/restore
- Support English and Chinese language
- Firmware upgrade by TFTP
- Syslog and CDR
- Web & telnet login password modify
- Factory reset
- Automatic time synchronization
- SNMP
- IVR local maintenance management

Dimension & Weight:

- 4FXS: 194(L)*133(W)*37(H)mm 0.69kg
- 8FXS: 242(L)*152(W)*40(H)mm 1kg

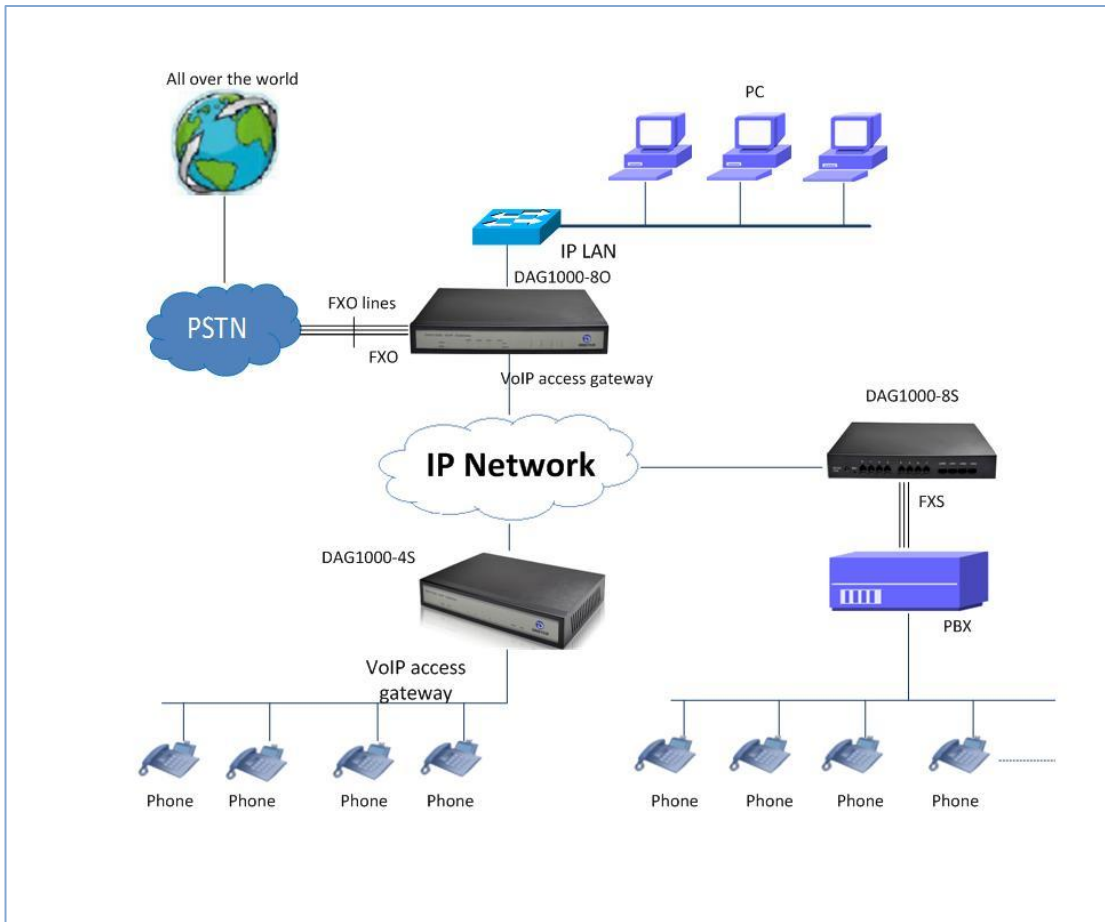
Packing Content

- Gateway
- Power adapter
- RJ-45 Network cable
- User's Manual CD



Trust Growth Cooperation Sharing

System Application



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The Gateway to VoIP World