

DAG1000-4S4O Hybrid Analog VoIP Gateway

Overview

DAG1000 series hybrid analog gateway are versatile IP-based voice and fax gateway. For small business, the family office, remote office and branch businesses provide a low cost, simple operation VoIP solutions. FXS/FXO hybrid gateway support network failure and power failure lifeline feature , flexibly realize interoperability with simulation PBX and provide reliable voice quality assurance for the traditional voice transition to IP voice. Support standard SIP protocol and compatible mainstream IPPBX and soft switch platform.

Pictures



Front

Key Features

- 4 FXS and FXO hybrid
- Flexible port group
- IP trunk
- Flexible routing and manipulation
- Data/ Voice/ Management VLAN
- · Voice and FAX all in one
- · Power and network failure lifeline
- · Firmware update automatically /configure update



Back



Physical Interfaces

Phone Interfaces: 4FX and 4FXO, RJ-11 Ethernet Interfaces: 1*WAN, 2*LAN,10/100Mbps

FXS

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

FXO

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

Voice & FAX

G.711A/U law, G.723.1, G.729A/B, Silence Suppression Comfort Noise Generation(CNG) Voice Activity Detection(VAD) Echo Cancellation(G.168), with up to 128ms Adaptive (Dynamic) Jitter Buffer Programmable Gain Control T.38/Pass-through Modem/POS DTMF mode: Signal/RFC2833/INBAND VLAN 802.1P/802.1Q

Layer3 QoS and DiffServ

VolP

Protocol:

SIP v2.0 (UDP/TCP),RFC3261 SDP,RTP(RFC2833), RFC3262, 3263,3264,3265,3515,2976,3311 SIP TLS/SRTP RTP/RTCP, RFC2198, 1889 RFC4028 Session Timer RFC3266 IPv6 in SDP RFC2806 TEL URI RFC3581 NAT,rport Outbound Proxy DNS SRV/ A Query/NATPR Query SIP Trunk Early Media/Early Answer NAT:STUN, Static/Dynamic NAT

Maintenance

SNMP v1/v2/v3 TR069 Auto Provisioning Web/Telnet Configuration Backup/Restore Firmware Upgrade via Web CDR Syslog Network Capture NTP/Daylight Saving Time IVR local Maintenance Cloud-based Management

Network

Static/Dynamic IP, PPPoE DHCP Client IPv4/IPv6 TCP, UDP,TFTP, FTP, ARP,RARP, Ping, NTP, SNTP, HTTP/HTTPS, DNS Ping / Tracert DHCP Option 66,120,121

Software Features

Port Group Web ACL Telnet ACL Action URL Digitmap Routing Rules based Prefixes Caller/Called Number Manipulation

Environmental

Power supply: Input 100~240VAC Output 12VDC, 2A Power consumption(W): 20W Operator Temperature: 0 °C ~ 45 °C Storage Temperature: -20 °C ~80 °C Humidity: 10%-90% no condensation Compliance: CE, FCC

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About Us

countries.

Founded in 2011 in Shenzhen, DINSTAR is a leading global provider of IP Unified Communication products including VoIP Gateways, IP PBXs, IP Phones and SBCs, we have been delivering more agile, efficient and affordable communication solutions and unparalleled communication experiences to our customers with our reliable, innovative and future-proof products for years. Through our value-added distributors and resellers worldwide, now DINSTAR serves telecom

operators, service providers, system integrators, enterprises, SMBs and OEM partners in over 100