

# DAG1000-1/2S Analog Telephone Adapter

## Overview





DAG1000-1S





DAG1000-2S

DAG1000-1/2S is a multi-functional analog telephone adapter which offering seamless connectivity between IP-based telephony networks and legacy telephones (POTS), fax machines and PBX systems. The device offers 1 or 2 FXS ports, fax over IP and integrated with high speed NAT router. These powerful ATAs features exceptional voice quality in various application environments such as SOHO, home users etc. Moreover, with automated provisioning and Centralized manage system, users can easily maintenance and volume deployment.

# **Key Features**

- Cost effective ATA with 1 or 2 FXS ports
- Fax over IP (T.38 and Pass-Through)
- Support IPv4 and IPv6
- TR069 and SNMP
- Elastix/Broadsoft Certification
- Multiple codecs: G.711A/U,G.723.1,G.729A/B, G.726
- Fully compatible with leading IMS/NGN, SIP based IP telephony system

# **Honesty Concentration**

## **Physical Interfaces**

Capacity

DAG1000-1S: 1\* FXS, RJ11 DAG1000-2S: 2\* FXS. RJ11

Ethernet Interfaces:

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

Console

N/A

### **Voice & FAX**

- G.711A/U law, G.723.1,
  G.729A/B,G.726
- Silence Suppression
- Comfort Noise Generation(CNG)
- Voice Activity Detection(VAD)
- Echo Cancellation(G.168), with up to 128ms
- Adaptive (Dynamic) Jitter Buffer
- Hook Flash
- Programmable Gain Control
- T.38/Pass-through
- Modem/POS
- DTMF modes:
  SIP Info/RFC2833/Inband
- VLAN 802.1P/802.1Q
- Layer3 QoS and DiffServ
- MOS: 4.192 (G.711 Codec)

#### **FXS**

Connector: RJ11

Dial Mode: DTMF and Pulse

Pulse: 10 and 20 PPS

Caller ID: DTMF/FSK CLI Presentation

Max Cable Length: 3KM

Reversed Polarity

Programmable Call Progress Tone

#### **VoIP**

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

# **Innovation Open-mind**

## **Supplementary Service**

- Call Waiting
- Blind Transfer
- Attend Transfer
- Call Forward on Busy
- Call Forward on No Reply
- Unconditional Call Forward
- Warm/Immediately Hotline
- Call Hold
- Do-not-disturb
- 3-Way Conference
- Message Waiting Indicator

### **Software Features**

- Hunting Group
- Web ACL
- Telnet ACL
- Action URL
- PPPoE/IPv4/IPv6
- Digitmap
- Bandwidth Optimization
- Routing Rules based Prefixes
- Caller/Called Number Manipulation

#### Maintenance

- SNMP v1/v2/v3
- TR069
- Auto Provisioning
- Web/Telnet
- Configuration Backup/Restore
- Firmware Upgrade via Web
- CDR
- Syslog
- Ping/Tracert Test
- Network Capture
- Outward Test(GR909)
- NTP/Daylight Saving Time
- IVR local Maintenance
- Cloud-based Management

### **Environmental**

Power Supply:

100-240VAC, 50-60 Hz@DC12V 1A

- Power Consumption:<5W</li>
- Operating Temperature:0 °C ~ 45 °C

Storage Temperature: -20 °C ~80 °C

- Humidity:10%-90% Non-Condensing
- Dimensions(W/D/H): 126\*76\*25mm
- Unit Weight: <=0.2kg</li>
- Compliance: UL

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