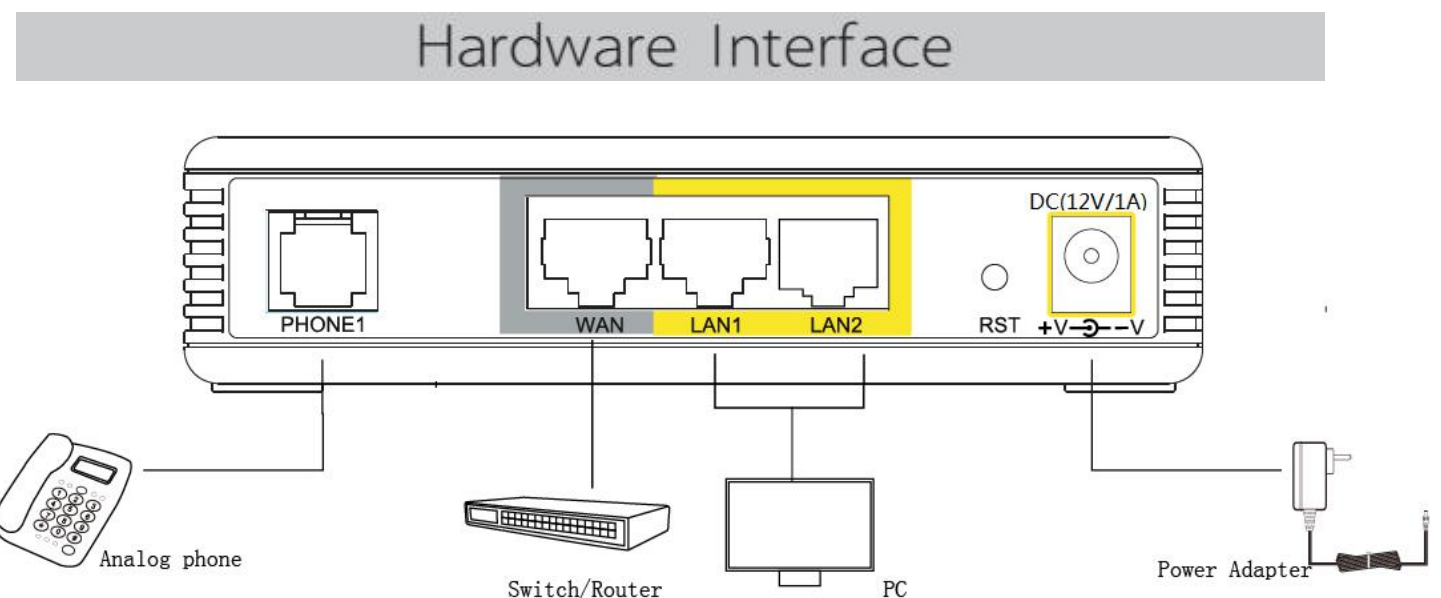


1 FXS Port VoIP Phone Adapter FTA5101



Feature Keys

- 1 FXS Port
- T.38 And T.30 FAX
- TR069,SNMP
- 10/100Mbps Ethernet



Overview

FTA5101, which has one FXS port , one LAN port and one WAN port, is one of the most popular VoIP ATAs researched and produced by FlyingVoice. This product can not only provide one SIP lines for users to make calls, but also it is a wire-speed NAT router, make you enjoy easy network atmosphere. What's more, FTA5101 support T.38 real time FAX and T.30 FAX with G.711.FTA5101 is a stand-alone device, which requires no PC to make Internet calls. This ATA guarantees clear and reliable voice quality on Internet, which is fully compatible with SIP industry standard and able to interoperate with many other SIP devices and software on the market.The FTA5101, 1 FXS Analogue Telephone Adapter products in Flyingvoice, which enables customers to register to different SIP Proxy

server, IP PBX and establish up to 1 concurrent VoIP calls for more flexibility in the voice communication.heir compact size, excellent voice quality, packed feature functionality and best-in-class price-performance point enable consumers to maximize the power of IP voice and data connectivity.FTA5101 is based on SIP V2.0 standard and compatibility with most service providers.TR069 CPE management & monitoring protocols and a base stand for vertical positioning.

Technical Parameters

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| Flash | <ul style="list-style-type: none">16MB Flash |
| RAM | <ul style="list-style-type: none">32MB SDRAM |
| Power | <ul style="list-style-type: none">AC/DC Adapter<ul style="list-style-type: none">AC Input: 100~240V, 50~60HzDC Output: 12V, 1A |
| Operating System | <ul style="list-style-type: none">Linux 2.6.21 |
| I/O Interfaces | <ul style="list-style-type: none">3 RJ-45 for 10/100Ethernet Ports2 RJ-11 for FXS Port |
| Environmental | <ul style="list-style-type: none">Operation Temperature: 0~50 Degree CStorage Temperature: -25~ 85 Degree CRelative Humidity: 10%~90% No Condensing |
| Audio Codec | <ul style="list-style-type: none">G.711(A/u),PAMS>4.3G.729A/AB,PAMS>4.0T.30 FAX with G.711Real time FAX over IP via T.38(Phone 2 port only)Adaptive Jitter BufferVoice Activity DetectionComfort Noise GenerationEcho Cancellation |
| Management | <ul style="list-style-type: none">Firmware UpgradeableWeb Management InterfaceIVR-driven Management InterfaceLocal and Remote Syslog (RFC3164)Auto ProvisioningSNTP Time SynchronizationMulti User LevelSupport IPv6SNMPv2TR069 |
| Protocols | <ul style="list-style-type: none">SIP V2 (RFC 3261)Backward Compatible with RFC2543Session Timer (RFC4028)SDP (RFC2327)RTP/RTCP (RFC1889 and RFC1890)NAPTR for SIP URI Lookup (RFC2915)STUN (RFC 3489)ARP/RARP (RFC 826/903)SNTP (RFC 2030)DHCP/PPPoEPPTP/L2TP VPNHTTP Server for Web ManagementTFTP/HTTP/HTTPS for Auto ProvisioningDNS/DNS SRV (RFC1706 and RFC 2782) |

| | |
|-----------------|--|
| Applications | <ul style="list-style-type: none">• NAT/NAPT Router function• MAC Address Cloning• DHCP Server• PPTP/L2TP VPN• PPPoE• SIP proxy redundancy<ul style="list-style-type: none">Dynamic via DNS SRV, A records• NAT Traversal by STUN• DMZ• QoS with Layer 3• DHCP Client and DHCP Server• IP conflict detectionv |
| Call Features | <ul style="list-style-type: none">• 3-way Conference• Music on hold• DTMF mode: In-band, RFC2833 and SIP INFO• Call Hold• Call Forwarding• Call Mute• Call Transfer• Call Waiting• Speed Dial• Caller ID and CWCID• Hotline• Real time fax over IP via T.38• T.30 FAX with G.711• Dial Plan• Black List• Call Log |
| SIZE And Weight | <ul style="list-style-type: none">• 120mm(L)x80mm(W)x34mm(H)• 116g(N.W) |