

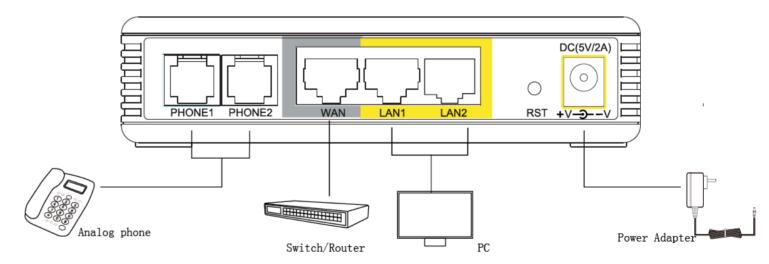
## 2 FXS Ports VoIP Adapter FTA5102



## **Feature Keys**

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

Hardware Interface



## **Overview**

FTA5102, which has two 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 two SIP lines for users to make calls, but also it is a wire-speed NAT router, make you enjoy easy network environment. What's more, FTA5102 support T.38 real time FAX and T.30 FAX with G.711.FTA5102 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 FTA5102, 2 FXS Analogue Telephone Adapter products in Flyingvoice, which enables customers to register to different SIP Proxy server, IP PBX and establish up to 2 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.FTA5102 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**

Power	AC/DC Adapter
	AC Input: 100~245V, 50~60Hz
	DC Output: 12V, 1A
Operating System	• Linux 2.6.36
I/O Interfaces	3 RJ-45 for 10/100Ethernet Ports
	2 RJ-11 for FXS Port
Environmental	Operation Temperature: 0~50 Degree C
	Storage Temperature: -25~ 85 Degree C
	<ul> <li>Relative Humidity: 10%~90% No Condensing</li> </ul>
Audio Codec	• G.711(A/u)
	• G.729A/AB
	• G726-32
	OPUS standard
	• T.30 FAX with G.711
	<ul> <li>Real time FAX over IP via T.38 (up to 14.4Kbps Pass-through)</li> </ul>
	Adaptive Jitter Buffer
	Voice Activity Detection
	Hook Flash
	Comfort Noise Generation
	Echo Cancellation
Management	Firmware Upgradeable
Management	Web Management Interface
	IVR-driven Management Interface
	Local and Remote Syslog (RFC3164)
	Auto Provisioning
	SNTP Time Synchronization
	Multi User Level
	Support IPv4, IPv6
	• SNMPv1/v2/v3
	• TR069
	FCMS cloud server
Protocols	<ul> <li>SIP V2 (RFC3261,RFC3262,RFC3263,RFC3264,RFC3265,RFC3515,RFC3891,RFC3892,3GPP,IMS)</li> </ul>
	Backward compatible with RFC2543
	Session timer (RFC4028)
	• SDP (RFC2327)
	• RTP/RTCP (RFC1889/RFC1890)
	UDP/TCP/TLS
	• IPv4/ IPv6
	<ul> <li>Support Dual Stack (IPv4 and IPv6)</li> </ul>
	NAPTR for SIP URI lookup (RFC2915)
	• STUN (RFC 3489)
	• ARP/RARP (RFC 826/903)
	• SNTP (RFC2030)

Voide over ap	FTAST02 Specific
	DHCP/IPoE /PPPoE
	PPTP/L2TP VPN
	HTTP Server for Web Management
	TFTP/HTTP/HTTPS for Auto Provisioning
	Message Waiting Indicator (RFC3842)
	DHCP Option Codes for SIP (RFC3361)
	DNS/DNS SRV (RFC1706 and RFC 2782)
	<ul> <li>VLAN mode: Voice/ Management VLANs, support VLAN 802.1p/ 802.1Q</li> </ul>
Applications	NAT/NAPT Router function
	MAC Address Clone
	DHCP Server
	PPTP/L2TP VPN
	• PPoE
	SIP proxy redundancy
	Dynamic via DNS SRV, A records
	NAT Traversal by STUN
	• DMZ
	QoS with Layer 3, Traffic shapping
	DHCP Client and DHCP Server
	IP conflict detection
Call Features	3-way Conference
	Music on hold
	Dial mode: DTMF and Pulse
	DTMF mode: In-band, RFC2833 and SIP INFO
	Call Hold
	Call Forwarding
	Call Mute
	Call Transfer
	Call Waiting
	Speed Dial
	Caller ID (Bellcore/DTMF/FSK) 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	• 120mm(L)x80mm(W)x32.5mm(H)
	• 119g(N.W)