

Partner Marketing Kit

HT813 *July, 2018*

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About the HT813

The HT813 is an analog telephone adapter that features 1 analog telephone FXS port and 1 PSTN line FXO port in order to offer backup lifeline support using a PSTN line. The integration of a FXO and FXS port enables this hybrid ATA to support remote calling to and from the PSTN line. For added flexibility, the FXS port extends VoIP service to one analog device. Users can convert their analog technology to VoIP thanks to the HT813's ultra-compact size, HD voice quality, advanced VoIP functionality, high-end security protection and multiple auto provisioning options. These advanced features also allow service providers to offer high quality IP service to customers looking to upgrade to VoIP.

<u>French</u> <u>Italian</u> <u>Russian</u>

<u>German</u> <u>Portuguese</u> <u>Spanish</u>

Messaging Strategy

HT813

Tagline	Hybrid ATA with FXS and FXO ports	
Subhead	Analog telephone adapter with backup lifeline support	
Positioning	Users looking to upgrade to or extend a VoIP network can turn to Grandstream's HT813. It offers HD voice quality, advanced VoIP functionality, high-end security protection and multiple auto provisioning options. The HT813 features 1 analog telephone FXS port and 1 PSTN line FXO port in order to offer backup lifeline support using a PSTN line. This ATA is ideal for small to medium sized deployments looking to upgrade to a VoIP network.	
Key Differentiators	 2 SIP accounts, 1 FXS port, 1 FXO port Dual 100Mbps LAN and WAN port Lifeline support 3-way voice conferencing Provisioning options Failover SIP server options 	

<u>French</u>	<u>ltalian</u>	<u>Russian</u>
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HT813 Datasheet



Hybrid ATA with FXS and FXO ports

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Supports 2 SIP profiles through 1 FXS port and 1 FXO



and WAN ports

Automated & secure Supports T.38 Fax for reliable Faxprovisioning options using TR069



Lifeline support (FXS port will be hardrelayed to FXO port) in case of power outage

 (\mathbf{X})



Failover SIP server automatically switches to secondary server if main server loses connection



3-way voice

conferencing per

Strong AES encryption certificate per unit

Telephone Interfaces One (1) RJ11 FXS port, One (1) RJ11 FXO PSTN line port with lifeline support Network Interface Two (2) 10/100Mbps ports (RJ45) with integrated NAT router LED Indicators POWER, LAN, WAN, FXS, FXO Factory Reset Button Yes Voice, Fax, Modem Telephony Features

Caller ID display or block, call waiting, flash, blind or attended transfer, forward, hold, do not disturb, 3-way conference Voice Codecs G.711 with Annex I (PLC) and Annex II (VAD/CNG), G.723.1, G.729A/B, G.726, iLBC, OPUS, dynamic jitter buffer, advanced line echo cancellation Fax over IP T.38 compliant Group 3 Fax Relay up to 14.4kpbs and auto-switch to G.711 for Fax Pass-through Short/Long Haul Ring Load 3 REN: Up to 1km on 24 AWG Caller ID Bellcore Type 1 & 2, ETSI, BT, NTT, and DTMF-based CID Disconnect Methods Busy Tone, Polarity Reversal/Wink, Loop Current Signaling Network Protocols TCP/IP/UDP, RTP/RTCP, HTTP/HTTPS, ARP/RARP, ICMP, DNS, DHCP, NTP, TFTP, TELNET, STUN, SIP (RFC3261), SIP over TCP/TLS, SRTP, TR-069 QoS Layer 2 (802.1Q VLAN, SIP/RTP 802.1p) and Layer 3 (ToS, Diffserv, MPLS) DTMF Method In-audio, RFC2833 and/or SIP INFO Provisioning and Control HTTP, HTTPS, TFTP, SSH, TR-069, secure and automated provisioning using AES encryption, syslog Security Media SRTP Control TLS/SIPS/HTTPS Management Syslog support, telnet, SSH, remote management using web browser Universal Power Supply Input: 100-240VAC, 50-60Hz Output: 12V/0.5A Operational: 32° – 104°F or 0° – 40°C Environmental Storage: 14° – 140°F or -10° – 60°C Humidity: 10 – 90% Non-condensing Dimension and Weight 130.5 x 90.5 x 29 mm (L x W x D) Weight 0.142Kg Compliance FCC/CE/C-TICK/ITU-K.21

English datasheet

French datasheet

German datasheet

Italian datasheet

Portuguese datasheet

Russian datasheet

Spanish datasheet

www.grandstream.com

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Marketing Assets

- 1. <u>HT813 pictures</u>
- 2. <u>Feature icons</u>
- 3. <u>Social Media banners</u>

- 4. Web banners
- 5. <u>Email banners</u>
- 6. <u>Licensed lifestyle images</u>