YEASTER



YEASTAR TG800 VOIP GSM GATEWAY

- 8 SIM Card Support
- SIP/IAX Protocol Support
- LAN 10/100Base-T Connectivity
- G.711a/u-law, G.722, G.726, G.729A, GSM, ADPCM, Speex Codecs
- SMS Center SMS Sending and Receiving
- Send Bulk SMS
- Call Detail Record [CDR]
- Firmware upgrade by HTTP/TFTP Gain Adjustment
- GSM/CDMA/UMTS Ports Group Manage Hotline
- Incoming /Outgoing Routing rules IP Blacklist
- Network Attack Alert NTP
- Send Bulk SMS Session Timer
- SIP Peer Mode: Support SIP Response Code Switch
- SIP server for IP phones: Support SIP Trunk: Suppor
- Open API for SMS and USSD
- · White List and Black List
- IP Blacklist

Read More

Price:

YEASTAR TG400 VOIP GSM GATEWAY

- Cost savings- save phone bills tremendously with mobile-tomobile calls and LCR
- Back up- work as a cost-effective backup when the landline network goes down
- Easy to install- everything can be easily set up in the web based management interface
- Easy to integrate- high compatibility with major IP PBX and softswitch brands
- SIP server and SIP trunk supported, SIP peer mode supported
- Calling type: VoIP to GSM, GSM to VoIP
- GSM ports group manage, VoIP trunk group
- Real Open API protocol (based on Asterisk)
- IP blacklist, Network attack alert, System logs

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YEASTAR TG200 VOIP GSM GATEWAY

- Number of GSM channels (Max): 2 (Quad-Band GSM/GPRS850/900/1800/1900MHz; Support external antenna)
- Protocol: SIP (RFC3261)
- Transport: UDP
- DTMF: RFC2833, SIP INFO, In-band
- Codec: G.711 A/u-law, G.726, G.729 A, GSM, Speex.
- LAN: 1 (10/100MBps)
- Network: DHCP Server, VLAN, Firewall.
- Power Supply: AC 100~240V/50~60Hz (DC 12V, 1A)
- Size: 175x125x31mm

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Price:

YEASTAR TG100 VOIP GSM GATEWAY

- Number of GSM channels (Max): 1
- Network type: 850/900/1800/1900MHz
- Protocol: SIP (RFC3261), IAX2
- Transport Protocol: UDP,TCP,TLS,SRTP
- DTMF: RFC2833, SIP INFO, In-band
- Codec: G.711 a/u-law, G.722, G.726, G.729a, GSM, ADPCM, Speex
- Echo Cancellation: ITU-T G.168 LEC
- LAN: 1 (10/100 Mbps)
- Network: Static IP, DHCP Client, Firewall, VLAN, DDNS, QoS, OpenVPN
- NAT Traversal: Static NAT, STUN
- Size: 110x70x24mm
- Power Supply: AC 100~240V/50~60Hz (DC 12V, 1A)







We offer On-premise IP PBX systems from Yeastar that are installed on site, where the phone system's functionality is programmed into the PBX hardware.

YEASTAR S50 VOIP PBX FOR 50 USERS

- Up to 50 users and 25 concurrent calls
- Includes 50 SIP trunks
- Add module cards to activate additional ports up to 8 Analog, 8 BRI, and 4 GSM
- Includes 5,000 minutes of expandable voicemail storage
- Built-in call recording capability

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Price:

We offer On-premise IP PBX systems from Yeastar that are installed on site, where the phone system's functionality is programmed into the PBX hardware.



YEASTAR S300 VOIP PBX

- 300 Users (expandable to 500)
- 60 Concurrent Calls (expandable to 120)
- Up to 24 FXS/FXO/BRI Ports
- Up to 6 GSM/3G/4G Channels
- Up to 3 E1/T1/PRI Ports

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We offer On-premise IP PBX systems from Yeastar that are installed on site, where the phone system's functionality is programmed into the PBX hardware.

YEASTAR S20 VOIP PBX SYSTEM FOR 20 USERS

- Up to 20 users and 10 concurrent calls
- Includes 20 SIP trunks
- Add module cards to activate additional ports up to 4 Analog, 4 BRI, and 1 GSM
- Includes 5,000 minutes of expandable voicemail storage
- · Built-in call recording capability
- Add on powerful apps including auto provisioning, LDAP, and more
- Free lifetime software/firmware upgrades
- Connect multiple sites and remote users instantly
- No hidden charges or extra licensing fees
- Low power consumption at 1.8-10.6W
- Automated attendant (IVR)
- Email to SMS/SMS to email
- Fax to email
- Paging/intercom
- Mobility extension
- · Music on hold
- SIP forking
- · Voicemail to email
- Desktop and wall-mount capable

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Price:

We offer On-premise IP PBX systems from Yeastar that are installed on site, where the phone system's functionality is programmed into the PBX hardware.



YEASTAR S100 VOIP PBX FOR 100 USERS

• Users: 100 (up to 200)

• Max Concurrent Calls: 30 (up to 60)

Max Analog Ports: 16
Max BRI Ports: 16
Max Cellular Ports: 6
Max E1/T1/J1 Ports: 2
VoIP Trunks: 100

Protocol: SIP (RFC3261), IAX2

Transport Protocol: UDP, TCP, TLS, SRTP

Codec: G.711 (alaw/ulaw), G.722, G.726, G.729A, GSM, Speex,

ADPCM, iLBC, H.263, H.263P, H.264, MPEG4
• DTMF: In-band, RFC4733, RFC2833, SIP INFO

LAN: 1 x 10/100/1000 Mbps
WAN: 1 x 10/100/1000 Mbps

• IP Services: Static IP, DHCP, VPN, Firewall, VLAN, DDNS, PPPoE, QoS,

Static NAT, STUN

• Voicemail: 10000 min (expandable)

Firewall: YesT.38 Fax: YesApp Center: Yes

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