Español | English | 简体中文

Call Yeastar via Skype

Products > Gateways > VoIP GSM Gateway

# NeoGate TG400 - VoIP GSM Gateway

### Reduce costs for SOHO and SMBs

NeoGate TG400 is a compact 4 channels VoIP GSM gateway that connects GSM network with VoIP. It is the ideal product for small and medium sized companies with heavy demands of calls to mobile networks. The cost-saving solution makes connection cheaper and efficient.

### Benefits

**1**) Cost Savings - Save phone bills tremendously with mobile-to-mobile calls and LCR.

**2**) Back up - Work as a cost-effective backup when the landline network goes down.

**3**) Easy to install - Everything can be easily set up in the Web based management interface.

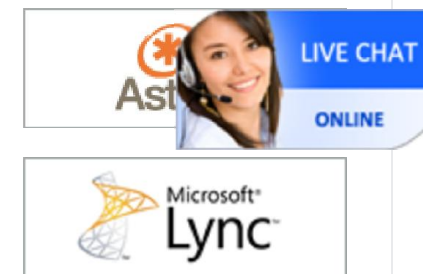**4**) Easy to integrate - High compatibility with major IP PBX and softswitch brands.

Show  NeoGate TG400

Product Material

📄 NeoGate TG400 Datasheet
📄 NeoGate TG400 Installtion Guide
📄 NeoGate TG400 User Manual

Applications

➡ Connect the IPPBX to GSM Network
➡ Call Saving
➡ Call Back

Compatible With

LIVE CHAT
ONLINE

Microsoft® Lync™

## ⇨ Specification

**Number of GSM channels (Max):** 4

**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 (10/100Mbps)

**Network:** Static IP, DHCP Client, Firewall, VLAN, DDNS,
QoS, OpenVPN

**NAT Traversal:** Static NAT, STUN


**Size:** 213x160x44mm

**Power Supply:** AC 100-240V 50/60Hz 0.5A MAX


**Operation Range:** 0° to 50°C, 32° to 122° F

**Storage Range:** -20° to 65°C, 4° to 149° F

**Humidity:** 10-90% non-condensing

## ⇨ Features

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

Incoming /Outgoing Routing rules

SMS Sending and Receiving

Send Bulk SMS

Gain Adjustment

USSD

PIN Modify

Carrier Selection: Auto/Manual

Balance Alarm

Caller ID/CLIR

Black List

Hotline

Call Duration Limitation

Call Transfer

Call Back

Call Status Display

Call Detail Record (CDR)

Call Progress Tone Generation

Call Duration Limitation for SIM Card/Single Call

LCR (Least Cost Routing)

Top voice quality (EFR super sound)

SIP Response Code Switch

Open API for SMS and USSD

Real Open API Protocol (Based on Asterisk)

IP Blacklist

Network Attack Alert

System Logs

Web based configuration

| **Products** | **Solutions** | **Support** | **Partner** | **About Us** |
| --- | --- | --- | --- | --- |
| IP Phone System | Phone system for SMBs | Downloads | Channel Partners | Profile |
|   - MyPBX SOHO | Remote Office Solution | FAQ | Technology Partners | Factory |
|   - MyPBX Standard | Redundancy Solution | Report a Problem | Become a Partner | News |
|   - MyPBX U Series | Billing System Solution | | | Fairs |
| NeoGate - VoIP Gateway | Make/Receive call via Skype | | | Contact Us |
| Recovision - NVR | Combine MyPBX with PABX | | | |
| FAX Server | | | | |

Contacts  |  Disclaimer  |  Privacy  |  Legal