

DBL Technology Voip/GSM gateways

- **Web site** : <http://www.dbltek.com/>

DBL Technology Ltd. is a leading company in developing and manufacturing advanced voice over IP (VoIP) solutions for the worldwide market. We are dedicated in the development of the latest VoIP technologies, and more than 10 years in developing a wide range of IP Phones and Gateways. The offering of high quality, innovation, affordability and flexibility inside each and every single one of our products is our commitment to you.

- **Installation** :

- Insert a SIM card and connect the Gateway to your router.
- Call the SIM number and enter *01 to know the IP address.
- Open your web browser and set the IP to access to the configuration interface (the default user/password are admin/admin).



- Set Basic VOIP settings

The screenshot shows the 'SIP' configuration page in the EasyPhone GoIP4 web interface. The interface has a dark blue header with the 'EasyPhone GoIP4' logo and '简体中文' (Simplified Chinese) text. A left sidebar contains a menu with categories: Status, Configurations, Preferences, Network, Basic VoIP, Advance VoIP, Media, Call Out, Call Out Auth, Call In, Call In Auth, SIM, SIM Forward, IMEI, SMS, GSM Carrier, GSM Base Station, and Tools. The main content area is titled 'SIP' and contains the following settings:

- Config Mode: Single Server Mode (dropdown)
- Phone Number: goip (text input)
- Display Name: (empty text input)
- Authentication ID: goip (text input)
- Password: **** (password input)
- SIP Proxy: (empty text input)
- SIP Registrar: 192.168.1.17 (text input)
- Re-register Period (s): 10 (text input)
- Outbound Proxy: (empty text input)
- Home Domain: (empty text input)
- Backup Server: Enable Disable
- Routing Prefix: Line1 Line2 Line3 Line4
- Line 1 Routing Prefix: (empty text input)
- *Auto Config Other lines (button)
- Save Changes (button)

- Set Call out settings

The screenshot shows the 'Call Out' configuration page in the EasyPhone GoIP4 web interface. The sidebar on the left lists various configuration categories, with 'Call Out' currently selected. The main configuration area includes the following settings:

- GSM Auto Redial:** Enable Disable
- GSM Dial Timeout(s):**
- Call OUT via GSM:** Enable Disable
- Forwarding to GSM Number:**
- Dial Plan:**
- Restricted Dial Plan:** Enable Disable
- Sleep Interval in between Calls (s):**
- Sleep Interval Mode:** Any Calls Answered Calls
- *Auto Config Other lines:**

A 'Save Changes' button is located at the bottom of the configuration area.

- **Configuration :**

SIP incoming trunk

```
[general]
Trunk Name: ovh-in (for example)

[Outgoing Settings]

Trunk Name: ovh-in (for example)
PEER Details:
  type=peer
  host=sipx.ovh.fr
  username=00339xxxxxxxx
  fromuser=00339xxxxxxxx
[Registration]

Register String: 00339xxxxxxxx:password@sipx.ovh.fr/00339xxxxxxxx
```

SIP outgoing trunk

[general]

Trunk Name: ovh-out (for example)

[Outgoing Settings]

Trunk Name: ovh-out (for example)

PEER Details:

type=peer

host=sipx.ovh.fr

username=00339xxxxxxxx

fromuser=00339xxxxxxxx

From:

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