



# DINSTAR DAG1000-40 Analog VoIP Gateway Installation Guide

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**DINSTAR DAG1000-40 Analog VoIP Gateway**



**Product Information**

The Dinstar VoIP Gateway is a telecommunications device that allows for the conversion of voice calls between traditional telephone lines and internet-based voice services. It comes in various models, including DAG1000-4O, DAG1000-8O, DAG2000-16O, DAG1000-4S4O, and DAG2000-8S8O. Each model has different ports and features to meet different connectivity requirements.

**Models and Ports**

Port Type	Model	WAN Ports	LAN Ports	FXS Ports	FXO Ports
WAN	DAG1000-4O	1	2	–	4
WAN	DAG1000-8O	1	2	–	8
WAN	DAG2000-16O	1	3	–	16
WAN + LAN	DAG1000-4S4O	1	2	4	4
WAN + LAN	DAG2000-8S8O	1	3	8	8

**Description of Indicators**

Indicator	Status	Description
PWR	On	The gateway is powered on.
PWR	Slow Flashing	The gateway is powered off or there is no power supply.
RUN	Fast Flashing	The gateway is running improperly.
RUN	On	The gateway is running properly.
FXS	On	The FXS port is idle or faulty.
FXS	Off	The FXS port is currently occupied by a call.
FXO	On	The FXO port is idle or faulty.
FXO	Off	The FXO port is currently occupied by a call.
WAN/LAN	Green Flashing	The gateway is not connected to the network or the network connection is improper.
WAN/LAN	On	The gateway is properly connected to the network.
WAN/LAN	Off	The network speed is set to work at 10Mbps.
WAN/LAN	Off	The network speed is set to work at 100Mbps.

## Installation Attentions

- Anti-jamming: To reduce interference with telephone calls, it is highly recommended to place the telephone lines connected to the gateway away from power cables.
- Power supply: DAG1000-4O, DAG1000-8O, and DAG1000-4S4O models require a 12VDC power adapter. DAG2000-16O and DAG2000-8S8O models accept AC input voltage of 100-240V. Ensure a safe and stable power supply.
- Network bandwidth: Ensure sufficient network bandwidth to guarantee stable operation of the gateway.
- Temperature and humidity: Install the gateway in an equipment room with appropriate temperature and humidity to avoid malfunctions.
- Ventilation: Avoid overheating by not piling up the gateway with other devices and ensuring good ventilation around the gateway.
- Mechanical load: Place the gateway steadily to avoid damage. It is advised to horizontally place the gateway on a flat surface or a cabinet.

THANKS FOR CHOOSING DINSTAR'S VOIP GATEWAY!

Please read this guide carefully before installing the gateway. If you need any technical support, please contact us.

**Tel:** +86 755 61919966

**Email:** [support@dinstar.com](mailto:support@dinstar.com)

**Web:** [www.dinstar.com](http://www.dinstar.com)

## Models and Ports

Port Type Model	WAN	LAN	FXS	FXO
DAG1000-4O	1	2	/	4
DAG1000-8O	1	2	/	8
DAG2000-16O	0	3	/	16
DAG1000-4S4O	1	2	4	4
DAG2000-8S8O	0	3	8	8

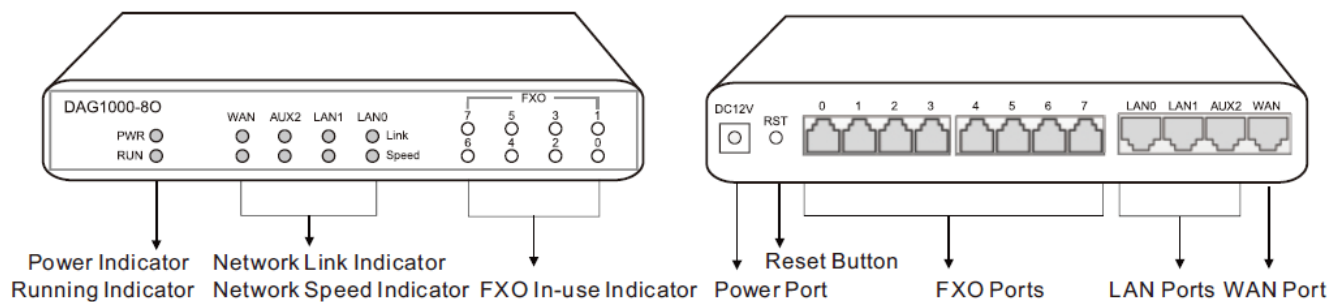
Please confirm your product model before installation and follow the installation guideline for the model

## Description of Indicators

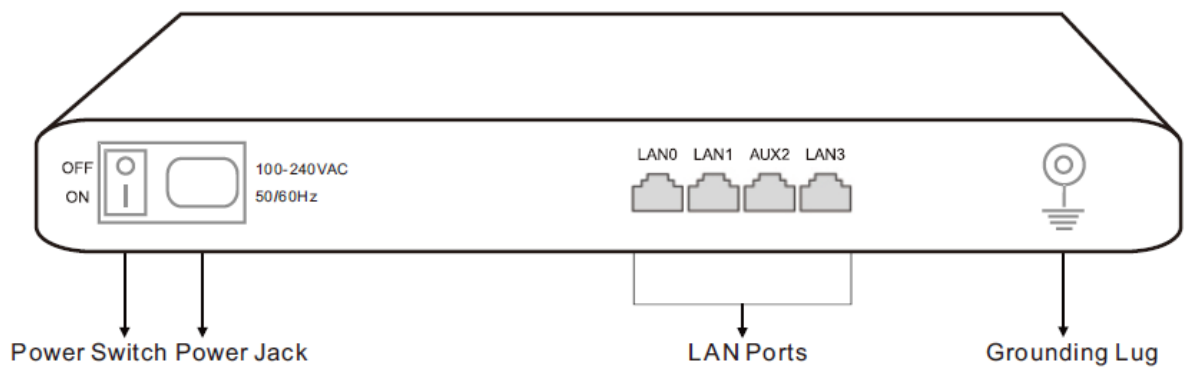
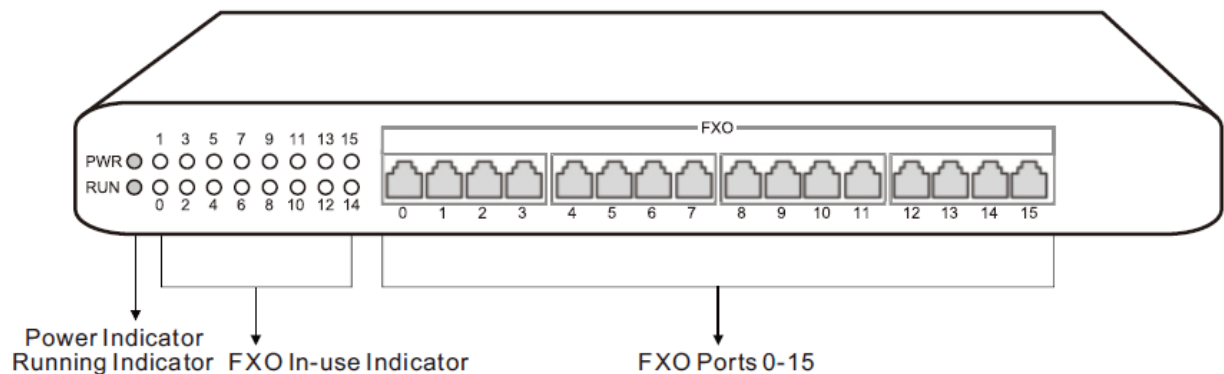
Indicator	Definition	Status	Description
PWR	Power Indicator	On	The gateway is powered on
		Of f	The gateway is powered off or there is no power supply
RUN	Running Indicator	Slow Flashing	The gateway is running properly
		Fast Flashing	If there is a SIP account registered successfully, the indicator is fast flashing
		Of f	The gateway is running improperly
FXS	Telephone In-use Indicator	On	FXS port is currently occupied by a call
		Of f	FXS port is idle or faulty
FXO	FXO In-use Indicator	On	FXO port is currently occupied
		Of f	FXO port is idle or faulty
WAN/LAN	Network Link Indicator	Green Flashing	The gateway is properly connected to network
		Of f	The gateway is not connected to network or network connection is improper way
	Network Speed Indicator	On	Work at 100Mbps
		Of f	Work at 10Mbps

## Indicators and Ports

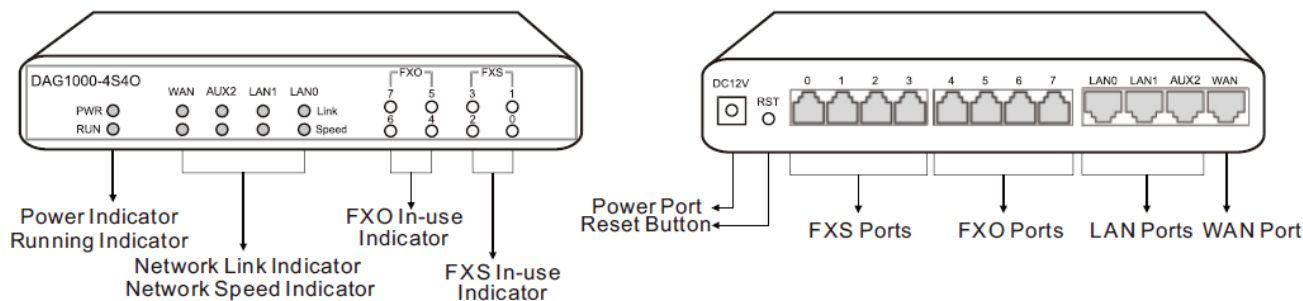
**DAG1000-4O/DAG1000-8O**



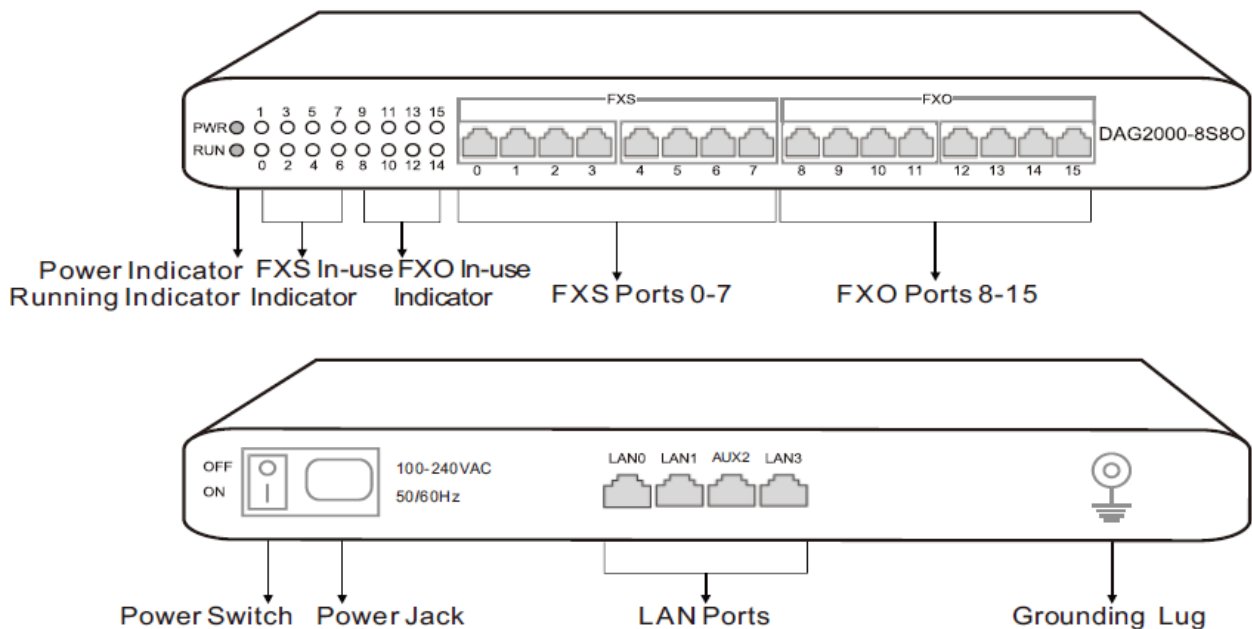
**DAG2000-16O**



**DAG1000-4S4O**



**DAG2000-8S8O**



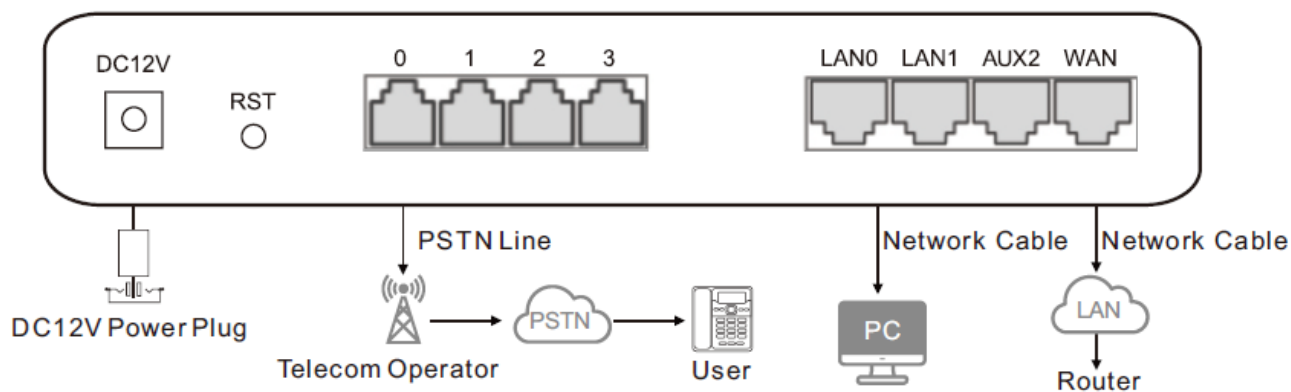
## Installation Attentions

- Anti-jamming: to reduce interference with telephone calls, it's highly recommended that telephone lines connected to the gateway should be placed away from power cables.
- Power supply: DAG1000-4O, DAG1000-8O and DAG1000-4S4O are equipped with 12VDC power adapter, while DAG2000-16O and DAG2000-8S8O accepts AC input voltage of 100-240V. Please ensure safe and stable power supply.
- Network bandwidth: please ensure there is enough network bandwidth so as to guarantee stabilized running of the gateway.
- Temperature and humidity: to avoid any accident that might cause malfunction, it's advised to install the gateway in an equipment room where temperature and humidity are appropriate.
- Ventilation: to avoid overheating, please do not pile up the gateway with other devices and make sure the gateway has good ventilation around.
- Mechanical load: please make sure the gateway is placed steadily to avoid damage. It's highly advised to horizontally place the gateway on a flat surface or a cabinet.

## Installation Instructions

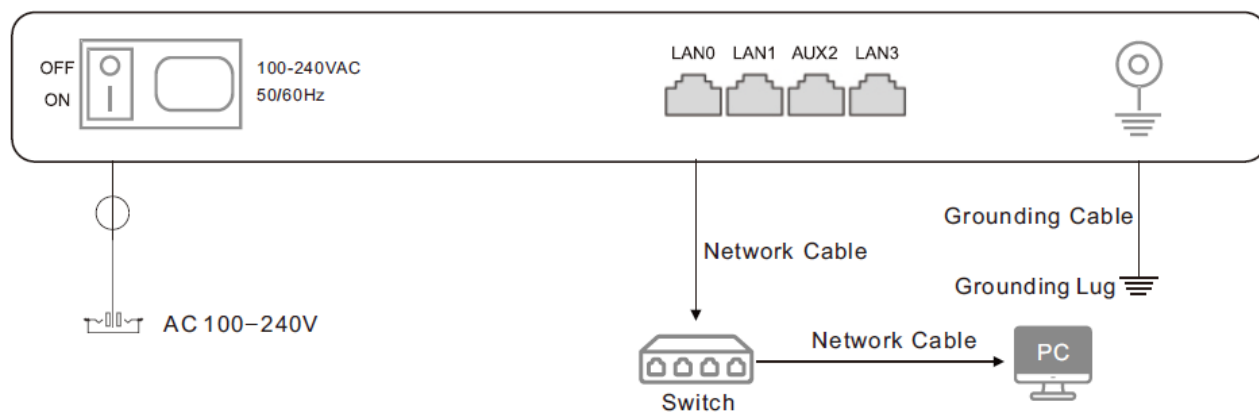
Connection Diagram for DAG1000-4O/DAG1000-8O Connect gateway with network, and connect gateway with power input. Connect FXO port with PSTN line.

•

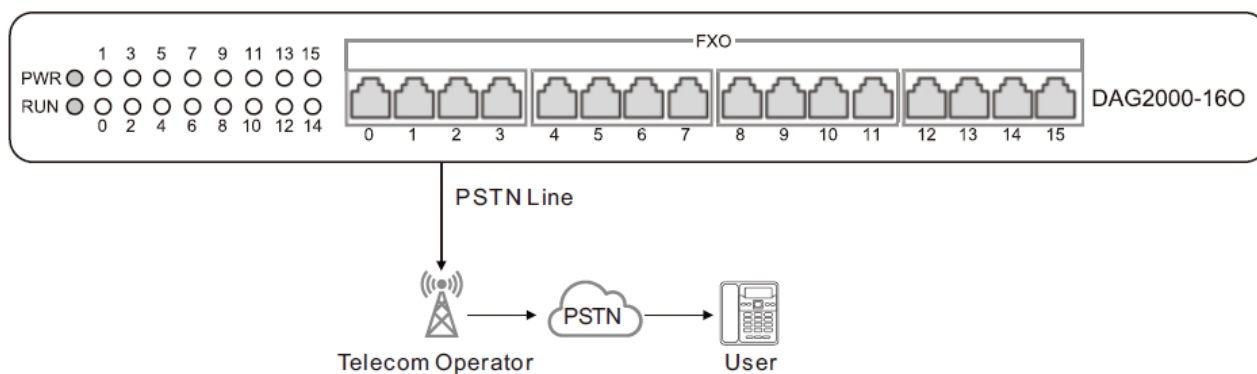


### Connection Diagram for DAG2000-16O

- Connect gateway with network, and connect gateway with power input.
- 

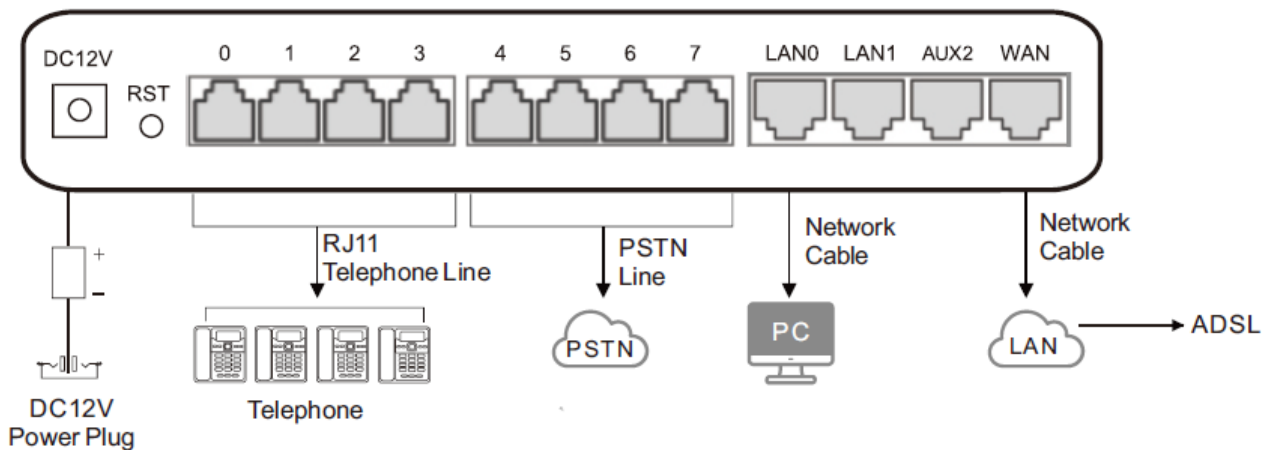


Connect FXO port with PSTN line.



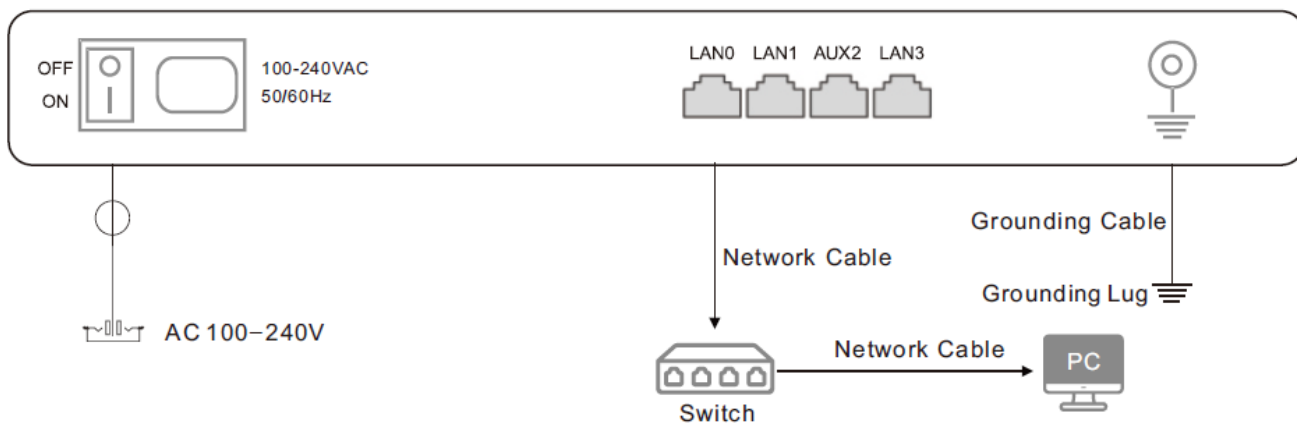
### Connection Diagram for DAG1000-4S4O

Connect gateway with network, and connect gateway with power input. Connect FXS port with Telephone line, and connect FXO port with PSTN line.

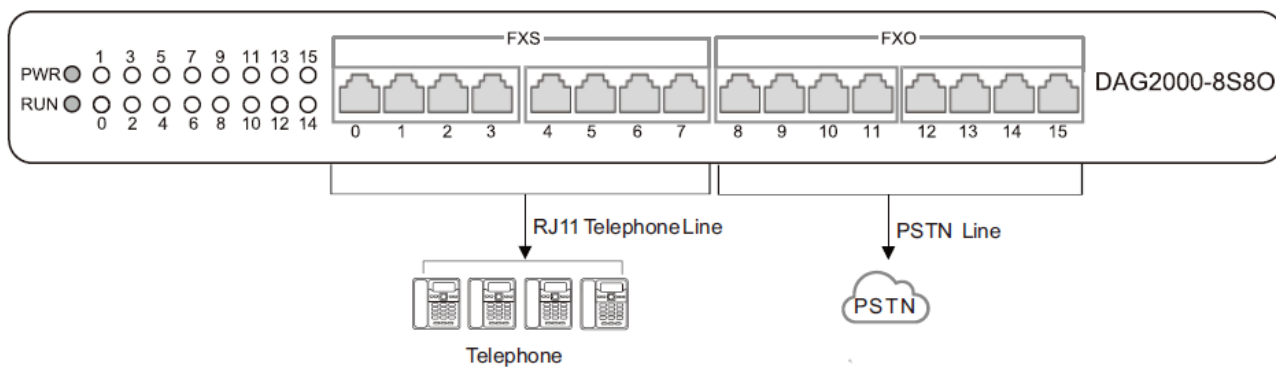


## Connection Diagram for DAG2000-8S80

- Connect gateway with network, and connect gateway with power input



- Connect FXS port with Telephone line, and connect FXO port with PSTN line.

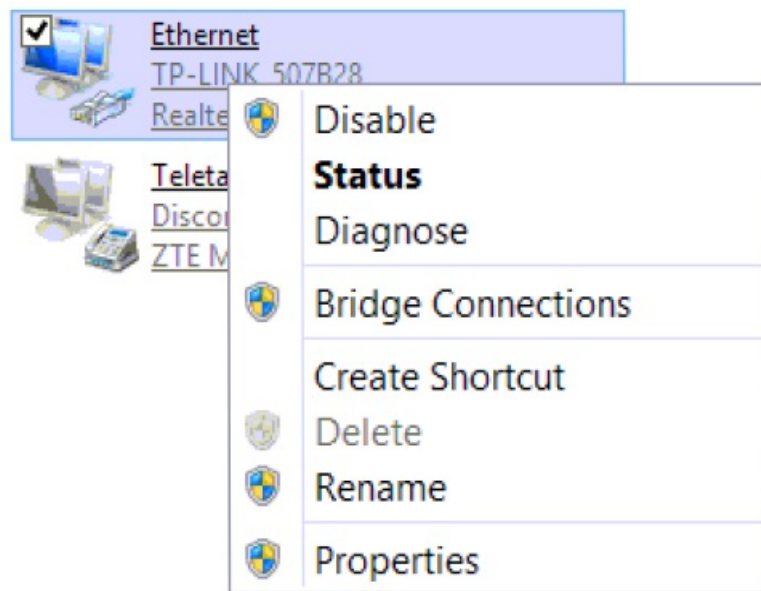


## Modify PC's IP Address

To log in the Web Management System of the gateway, firstly, you need to modify the IP address of PC which is used to access the gateway and to make it at the same network segment with the gateway.

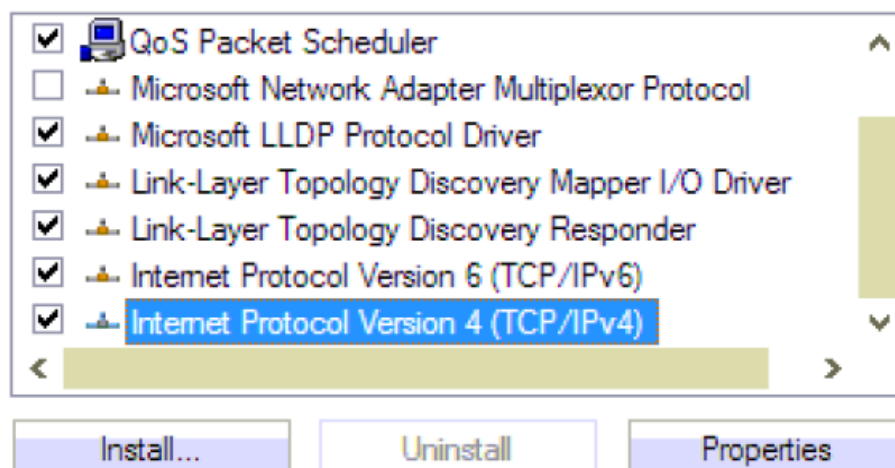
1. On the PC, click 'Network (or Ethernet) → Properties'.



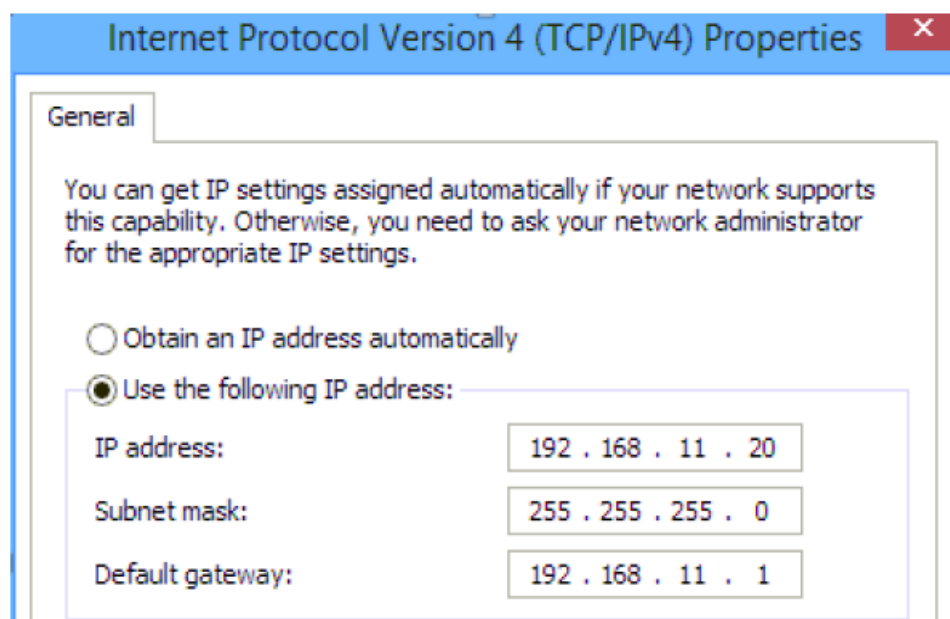


2. Double-click 'Internet Protocol Version 4 (TCP/IPv4)'.

This connection uses the following items:



3. Select 'Use the following IP address', and then enter an available IP address '192.168.11.XXX' which is at the same network segment with '192.168.11.1'



**Log in Web Management System**

Open a web browser and enter the gateway's IP address (default IP is 192.168.11.1).

Then input username and password in the displayed login GUI. Both the default username and password are admin.

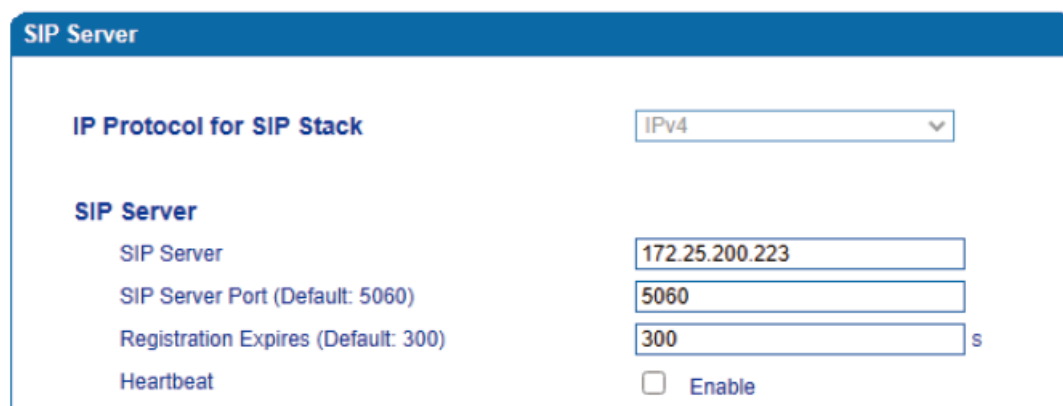
## Modify Gateway's IP Address

If you want to modify the IP address of the DAG gateway, please click 'Network → Local Network' in the navigation tree on the left of the Web Management System, and then input new IP address and DNS server address on the displayed interface

## Basic Configurations of FXS

### Configure SIP Server

Log in the Web Management System of the gateway, and then click SIP Server in the navigation tree and then enter the address of the SIP server which is to be registered. Click Save in the last.



The screenshot shows the 'SIP Server' configuration page. At the top, there's a blue header with the text 'SIP Server'. Below it, the 'IP Protocol for SIP Stack' is set to 'IPv4' in a dropdown menu. Under the 'SIP Server' section, there are four input fields: 'SIP Server' with the value '172.25.200.223', 'SIP Server Port (Default: 5060)' with the value '5060', 'Registration Expires (Default: 300)' with the value '300', and 'Heartbeat' with an unchecked checkbox and the label 'Enable'.

SIP Server	
IP Protocol for SIP Stack	IPv4
<b>SIP Server</b>	
SIP Server	172.25.200.223
SIP Server Port (Default: 5060)	5060
Registration Expires (Default: 300)	300 s
Heartbeat	<input type="checkbox"/> Enable

### Configure SIP Account Information

Click Port in the navigation tree on the left, and then click Add. Input the SIP account and authentication password assigned by the SIP server.

Port Add

Port	0
Disable Port	<input type="checkbox"/>
Registration	<input checked="" type="checkbox"/> Enable
IP Profile	0 <default>
Tel Profile	0 <default>
Display Name	Amy
SIP User ID	201
Authenticate ID	201
Authenticate Password	*****
Offhook Auto-Dial	
Auto-Dial Delay Time	s
DND(Do Not Disturb)	<input type="checkbox"/> Enable
Caller-ID	<input checked="" type="checkbox"/> Enable
Number for CFU(Call Forwarding Unconditional)	
Number for CFB(Call Forwarding Busy)	
Number for CFNRy(Call Forwarding No Reply)	
Call Waiting	<input type="checkbox"/> Enable
Play Call Waiting Tone	<input type="checkbox"/> Enable
Call Waiting Send CID	<input type="checkbox"/> Enable

Click Save and then restart the gateway for the new configurations to take effect.

## Basic 10 Configurations of FXO

### Configure Port Group

You are advised to divide PSTN lines into several groups and then distribute them to different port groups. Generally, PSTN lines belonging to a same telecom operator are distributed to a same port group. Click Call & Routing→Port Group, and then click Add on the displayed interface

Port Group Add

Index	1
Registration	<input checked="" type="checkbox"/> Enable
IP Profile	0 <default>
Description	TELE
Display Name	TELECOM
SIP User ID	10000
Authenticate ID	10000
Authenticate Password	
Offhook Auto-Dial	
Auto-Dial Delay Time	
Port Select	Cyclic Ascending
Pick Up on Group	*#
Call Answer Timeout	15
Select Port Count	Cyclic Select
Port	Select Port for this Group

When calls come from PSTN to the gateway, the off-hook auto-dial number is generally regarded as the DID (Direct-inward-dial) number or destination number of the SIP server.  
Click Select Port for this Group, and the following interface will be displayed, and then click Save to take effects.

## Configure IP Trunk

On the Call & Routing→IP Trunk interface, click Add to create an IP trunk which connects to a peer SIP server.

IP Trunk Add

Index	63
Description	eLA
Remote Address	10.10.0.50
Remote Port	5060
Heartbeat	<input type="checkbox"/> Enable

## Create Outbound Route

On the interface, Call & Routing→<IP→Tel Routing> click Add to create an outbound route for the IP trunk. As shown in the following figure, Caller/Callee Prefix is set as 'any' and calls from IP trunk 63<eLA> are routed to port group 1<TELE>.

**IP->Tel Routing Add**

Index	127	
Description	outgoing call	
Calls from	<input checked="" type="radio"/> IP Trunk <input type="radio"/> SIP Server	63 <eLA>
Caller Prefix	any	
Called Prefix	any	
Calls to	<input type="radio"/> Port <input checked="" type="radio"/> Port Group	1 <TELE>

## Create Inbound Route

On the Call & Routing→<Tel→IP/Tel Routing> interface, click Add to create an inbound route for the IP trunk. As shown in the following figure, Caller Prefix is set as 'any' and calls from port group 1<TELE> are routed to IP trunk 63<eLA>.

## Configure FXO Parameters

On the Advanced →FXS/FXO interface, configure the gateway to detect CID after ring or before ring, and enable 'Send Original CID when call from PSTN'. Format of "from" field is set as 'CID/CID' when CID is available, while format of "from" field is set as 'Display/User ID' when CID is unavailable.

**FXO Parameter**

FXO Concurrent Calls(0 means unlimited)	0
Incoming Call from PSTN	
Configuration by FXO	<input checked="" type="checkbox"/> Enable
Detect CID	Before Ring
Obtain FSK CID from	Num
Send Original CID when Call from PSTN	<input checked="" type="checkbox"/> Enable
Format of "From" field when CID is Available	CID/CID
Format of "From" field when CID is Unavailable	Display/User ID
CID : Calling Number	
FXO Keep Onhook until Called Answered(Need Enable Auto-Dial)	<input checked="" type="checkbox"/> Enable
Interval of Offhook and Onhook When Called Rejected	600 ms
Allow Call to SIP Server without Registration	<input checked="" type="checkbox"/> Enable
Ignore Call when SIP Unregistered	<input type="checkbox"/> Enable

## Configure Priority of Codecs

On the Advanced→Media Parameter interface, configure the priority of preferred codecs. The gateway supports G.729, G.723, G.711U and G.711A.

## Configure Impedance Value

**Preferred Vocoder**

	Coder Name	Payload Type	Packetization Time(ms)	Rate(kbps)	Silence Suppression
1st	G711A	8	20	64	Enable
2nd	G711U	0	20	64	Enable
3rd	G723	4	30	6.3	Enable

- To improve the compatibility of FXO ports, the gateway provides auto test to get the most matched impedance value.
- On the Advanced→FXS/FXO interface, click Start Test and then the gateway will automatically detect the best impedance for FXO lines. After the test is complete, save the impedance value.

The screenshot shows a configuration window for the FXS/FXO interface. It contains the following elements:

- A label "DC Impedance" with a dropdown menu set to "50 Ohm".
- A label "AC Impedance" with a radio button selected for "Impedance Value" and another for "Country".
- A label "Impedance Value" with a dropdown menu set to "600 Ohm".
- A label "Impedance Auto Test" with a "Start Test" button.

## More Details

This document only provides instructions for quick installation and basic configuration, For detailed configuration and parameter explanation, please make reference to user manual or ask for online technical support.

## Basic Operations

- Dial \*158# to query the IP address of LAN port of the gateway;
- Dial \*159# to query the IP address of WAN port of the gateway;
- Dial \*114# to query the telephone number of a FXS port;
- Dial \*165\*000000# to restore default IP address, user name and password;
- Restore factory default settings:
  - dial \*166\*000000#;
  - press the RST button for 7 seconds;
  - Log in the gateway and then click 'Tools → Factory Reset', then click Apply and restart the gateway.
- Dial \*111# to restart the gateway.

## FAQ

1. What if the caller number is not displayed when a call comes from PSTN ?
  1. A. Consult the telecom operator about whether the CID function has been enabled, or connect the PSTN line with a telephone to check it;
  2. B. On the Advanced→FXS/FXO interface, check the gateway is configured to detect CID before ring or after ring, and then check whether "Send Original CID when Call from PSTN" is enabled or not.
2. Can the gateway be connected with PSTN lines from different telecom operators ? Yes, it can be realized by creating port groups to manage PSTN lines belonging to different telecom operators.

What if the gateway's IP address is forgotten ?

1. Dial the number of PSTN line first, and then dial \*158# to query the IP address of LAN port of the gateway.
3. How to restore factory default settings ?
    1. A. Press the RST button for 7 seconds;
    2. B. Dial the number of PSTN line first, and then dial \*166\*000000#. After that, restart the gateway.
    3. C. Log in the gateway and then click Tools → Factory Reset, then click Apply and restart the gateway.



Web: [www.dinstar.com](http://www.dinstar.com)

Documents / Resources

<div><p>Quick Installation Guide</p><ul style="list-style-type: none"><li># DAG1000-4O</li><li># DAG1000-4O</li><li># DAG2000-16O</li><li># DAG1000-884O</li><li># DAG2000-888O</li></ul><p>THANKS FOR CHOOSING DINSTAR'S VOIP GATEWAY!</p></div>	<p><a href="#">DINSTAR DAG1000-4O Analog VoIP Gateway</a> [pdf] Installation Guide</p> <p>DAG1000-4O Analog VoIP Gateway, DAG1000-4O, Analog VoIP Gateway, VoIP Gateway, Gate way</p>
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References

- [D VoIP Gateway | IP PBX | IP Phone Session Border Controller SIP Intercom DINSTAR](#)