

DVG-2101SP VOIP TELEPHONE ADAPTER

USER MANUAL





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Introduction

The D-Link DVG-2101SP VoIP Telephone Adapter links conventional telephony devices such as analog phones or fax machines to IP networks. The DVG-2101SP includes one Ethernet port and one RJ-11 telephone jack that provides voice communication over the IP network. It is also equipped with one PSTN (RJ-11) jack to allow the device to connect to the PSTN line. The Ethernet port is for a DSL/Cable Modem or other WAN (Wide Area Network) devices such as a Router or Gateway. It can be configured and monitored using a web browser.

This VoIP Telephone Adapter can reduce or eliminate long distance or inter-office phone charges by routing calls over the Internet or any IP network. Corporations can also enjoy the benefits of network consolidation and reduction of leased lines by relying on the Internet service providers to deliver toll-quality voice communications over the IP networks.

With this device, you can make and receive calls from the Internet or PSTN (make only).

Package Contents

- 1 x DVG-2101SP VoIP Telephone Adapter
- 1 x CAT5 Ethernet Cable (Blue)
- 2 x RJ-11 Phone Cables (Grey)
- 1 x 12V/1.5A AC/DC Adapter
- 1 x CD-ROM with Product Documentation
 - User Manual
 - Quick Setup Guide

Features

• WAN Support

- One 10/100Mbps auto-MDI/MDIX Ethernet port
- WAN Type: Static IP, Dynamic IP & PPPoE
- MAC Address Clone
- NAT Traversal: Outbound Proxy, STUN
- Dynamic DNS support: DLinkDDNS, DynDNS
- VLAN ID Tag
- QoS: IP Precedence
- NTP

• Management

- Web-based (HTTP), Telnet, and Telephone keypad configuration (dial prefix)
- Built-in Ping Tool
- Remote firmware upgrade via TFTP, HTTP, HTTPS, and FTP
- Multi-function reset button: IVR , reset , restore factory default settings
- Password Protected controlled Admin and User Access Authority
- Auto Provisioning:
 - + Web-based Administration and Configuration via Integrated Web Server
 - + Telephone Key Pad Configuration with Interactive Voice Prompts, Automated Provisioning & Upgrade via HTTPS, HTTP, FTP
 - Interactive Voice Response (IVR)
- SNMP V1/V2C

• Call In & Call Out

- Voice over IP Call
- PSTN Phone Call (out only)
- IP Address Calling
- Three Way IP to IP Conference Call

• Call Features

- Call Forwarding: No Answer / Busy / Unconditional
- Call Transfer: Unattended / Attended
- Call Waiting / Call Pickup / Call Hold / Call Back On Busy / Call Return
- Three Way Calling (Media Server required)
- MWI (Message Waiting Indicator) based on LED & Tone
- DND (Do Not Disturb)
- Warm Line
- Hot Line
- Selective / Anonymous Call Rejection
- Music On Hold
- Caller ID (Type I and Type II)
- Caller ID Blocking
- Speed Dialing, Repeat Dialing
- Consultation Hold

• Voice Functionality

- SIPv2 (RFC3261) compliance
- SIP Proxy Redundancy: Dynamic via DNS SRV, A Records Re-registration with Primary SIP Proxy Server
- SIP Extension: Session Timer, Proxy-Require
- MD5 Authentication for SIP
- SIP NAT Keep Alive Timer
- VPN
- Flash Hook Timer
- Adaptive Jitter Buffer
- Programmable Gain Control
- In-band DTMF
- Out-of-band DTMF relay: RFC2833 / SIP Info
- DTMF
- Termination Impedance: 600 / 900 & complex Impedance
- Failover SIP Proxy server registrations
- T.30 FAX pass through, T.38 real time FAX relay
- Caller ID: DTMF, FSK-Bellcore, FSK-ETSI detection and generation by country select
- Telephone book
- Voice Codec Feature: G.711 u-law/a-law, G.723.1/iLBC (optional), G.726, G.729.A/B
- VAD (Voice Activity Detection)
- CNG (Comfort Noise Generation)
- G.165/G.168 Echo Cancellation
- Silence Suppression & Detection

Front Panel

Register LED

This LED will light solid if the phone account registers successfully. This LED will flash during registering and light off if registering fails.

Power LED

This LED will light solid when the device finishes a self-test and booting up. This LED will flash during device self-test and booting and light off if the self-test fails.

Phone LED

This LED will light solid when the telephone is off-hook. A fast blinking LED indicates an incoming call and a slow blinking LED indicates a message is waiting. This LED will light off if the phone connected to the phone port is on-hook.



Provision LED

This LED will light up solid if the device provisions successfully. The LED will flash during provisioning and will be off if provisioning fails.

Line LED

This LED will light solid when a connection is established and blink to indicate activity.

WAN LED

This LED will light solid when a connection is established and blink to indicate activity. If this LED does not light up when a cable is connected, verify the cable connections and make sure your devices are powered on.

Back Panel



Hardware Installation

Connecting the DVG-2101SP Directly to a Modem in Bridge Mode

If your computer connects directly to a DSL or Cable modem and does not connect to a router, follow the steps below to install your DVG-2101SP.

Note: If your modem only has the one Ethernet port and it is being used by your computer then you will require a Router/Gateway, please see the next page for details on connecting using a Router/Gateway.



- 1. Turn off your computer.
- 2. Disconnect the power to your Cable/DSL modem (unplug or turn off the power switch).
- 3. Unpack the Ethernet cable (blue) that comes with DVG-2101SP. Attach one end of the Ethernet cable (blue) provided in this package to a LAN or Ethernet port on your Cable/DSL modem. If there are no Ethernet ports available on your modem, you should include a Router/Gateway in your network setup (see diagram on the next page).
- 4. Attach the other end of the provided Ethernet cable to the WAN port of the DVG-2101SP.
- 5. Attach one end of the provided phone cable (grey) to a standard analog telephone.
- 6. Attach the other end of the phone cable to the PHONE port on the rear panel of the DVG-2101SP.
- 7. Attach one end of the provided phone cable (grey) to a PSTN wall socket.
- 8. Attach the other end of the phone cable to the LINE port on the rear panel of the DVG-2101SP.
- 9. Reconnect the power to the Cable/DSL modem (plug in or turn on the power switch).
- 10. Unpack the 12V DC power adapter that comes with the DVG-2101SP. Connect the power adapter to the power connecter on the DVG-2101SP.
- 11. Connect the other end of the power adapter to an available electrical outlet (wall socket or surge protector).
- 12. Restart your PC.

Connecting the DVG-2101SP Behind a Router (Recommended)

If you wish to connect your DVG-2101SP behind a router, follow the steps below. After the steps are completed, your setup should look similar to the diagram below.



- 1. Turn off your computer.
- 2. Disconnect the power to the Cable/DSL modem (unplug the modem or turn off the power switch).
- 3. Disconnect the power to your existing router (unplug the router or turn off the power switch).
- 4. Unpack the Ethernet cable (blue) that came with the DVG-2101SP.
- 5. Attach one end of this Ethernet cable to the WAN port on the rear of the DVG-2101SP.
- 6. Attach the other end of this Ethernet cable to a LAN port on your existing router.
- 7. Be sure that the Ethernet cable that is plugged into the WAN port of your existing router is still plugged into your Cable/DSL modem.
- 8. Unpack the phone cable (grey) that came with the DVG-2101SP. Attach one end of this cable to a standard analog telephone.
- 9. Attach the other end of this cable to the PHONE port on the rear of the DVG-2101SP.
- 10. Attach one end of the provided phone cable (grey) to a PSTN wall socket.
- 11. Attach the other end of the phone cable to the LINE port on the rear panel of the DVG-2101SP.
- 12. Unpack the 12V DC power adapter that came with the DVG-2101SP.
- 13. Connect the power adapter to the power connecter on the DVG-2101SP.
- 14. Connect the other end of the power adapter to an available electrical outlet (wall socket or surge protector).
- 15. Reconnect the power to the Cable/DSL modem (plug in or turn on the power switch).
- 16. Reconnect the power to your existing router (plug in or turn on the power switch).
- 17. Restart your PC.

By default the unit is already configured to work behind a router. Be sure to change your computer's IP settings to DHCP (to automatically be assigned an IP address from the DVG-2101SP) regardless of what type of Internet connection you have. (Please refer to your Network Interface Card's manual for more detailed information.)

Configuration

Prior to using your Web Browser for accessing the web-based configuration utility, if DHCP is not enabled on your network, be sure to assign a static IP address to the VoIP Telephone Adapter.

Use the computer that was last connected directly to your Cable/DSL modem. Right-click My Network Places > select Properties > right-click Local Area Connection > select Properties > double-click Internet Protocol (TCP/IP).

Set your PC's IP address to 192.168.1.1 and the subnet mask to 255.255.255.0. The default gateway and primary DNS server IP addresses should be the LAN IP address of the	Internet Protocol (TCP/IP) Properties General You can get IP settings assigned automatically if your network supports this capability. Otherwise, you need to ask your network administrator for the appropriate IP settings. Obtain an IP address automatically IP address: IP address: IP address: Subnet mask: 255.255.255.0 Default gateway:
DVG-2101SP (192.168.1.150). Click OK .	Obtain DNS server address automatically OUse the following DNS server addresses: Preferred DNS server: Alternate DNS server: Advanced DK Cancel

In order to use a web browser to configure the VoIP Adapter you must make sure it has a valid Ethernet connection to a PC or LAN via its LAN port. Access the configuration utility to check the LAN port by entering the IP Address into your web browser address field.







Setup

Internet Setup

Connection Type

The WAN connection type setting allows you to configure this device to establish your Broadband Internet connection. Your Internet Service Provider (ISP) should provide the necessary information to you when you subscribe to the Internet access service, unless you are connecting the DVG-2101SP behind a Router / Gateway device.

Dynamic IP

Choose Dynamic IP if the VoIP Telephone Adapter is to obtain its WAN IP address from a DHCP server (e.g. Router / Gateway device).



Host Name

Input the host name of this device to be shown in teh DHCP server log.

Static IP

Choose Static IP Address if the WAN IP information is provided to you by your ISP or you want to assign a Static IP address to the DVG-2101SP manually. You will need to enter in the IP address, Subnet Mask, Gateway address, and DNS addresses. You should enter the IP address, Subnet Mask, and Gateway address in this page and enter the DNS addresses in the DNS page which is under the **Advanced** tab.

Product Page: DVG-2101SF	P			Site Map	Firmware Version: 1.00.378
D-Lin l	K				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
Wizard	WAN				Helpful Hints
Internet Setup VoIP Setup Time and Date Logout	The WAN connection to connection. Your Intern subscribe the Internet	ype setting allows you to configur net Service Provider (ISP) should access service. Protocol:	e this device to establish your provide the necessary inform tatic IP	broadband Internet ation to you when you	When configuring the D- Link VoIP device to access the Internet, be sure to choose the correct Connection Type from the list below. If you are unsure of which option to
	STATIC IP				choose, contact your Internet Service Provider (ISP).
		IP Address:		Optional)	If you are having trouble accessing the Internet through the D-Link VoIP device, double check any settings you have entered on this page and verify them with your ISP if
		Apply	Cancel		needed.
					More
BROADBAND					

IP Address Subnet Mask Gateway Input the IP Address you wish to assign to the WAN port.

Input your Subnet Mask. (All devices in the network must have the same subnet mask.) Input the Gateway IP address, usually the IP address of the Router or Gateway on your network.

PPPoE

Choose PPPoE (Point-to-Point Protocol over Ethernet) if your ISP uses a PPPoE connection. Your ISP will provide you with a username and password.

Product Page: DVG-2101SF	P			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lin	K				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
Wizard	WAN				Helpful Hints
Internet Setup	The WAN connection type	e setting allows you to configu	re this device to establish you	r broadband Internet	When configuring the D- Link VoIP device to access
VoIP Setup	subscribe the Internet ac	cess service.	provide the necessary inform	ation to you when you	the Internet, be sure to choose the correct
Time and Date		Protocol:	PPPoE 🔻		Connection Type from the list below. If you are
Logout					unsure of which option to choose, contact your
	РРРОЕ				Internet Service Provider (ISP).
		PPPoE User Name : PPPoE Password : Check Password : PPPoE Service Name : Idle Time Out : ✓ Auto Reconnect MTU : Apply)min ▼ 192 (64~1492) Cancel	(Optional)	If you are having trouble accessing the Internet through the D-Link VoIP device, double check any settings you have entered on this page and verify them with your ISP if needed. More
BROADBAND					

PPPoE User Name PPPoE Password Check Password	Enter the PPPoE user name supplied by your ISP. Enter the PPPoE password supplied by your ISP. Patype the password entered in the provious field
PPPoF Service Name	Enter the PPPoF Service Name (if supplied by your ISP)
Idle Time Out	Set the specific period of time, in minutes, to disconnect from the ISP when no traffic is passing through. The setting 0 minute means the VoIP adapter will never disconnect from the ISP.
Auto Reconnect	Check this checkbox to allow the VoIP Adapter to re-establish the connection if it is terminated by the ISP. However, if the connection is terminated under any other conditions (i.e. by Idle Timeout or manual disconnect), the VoIP Telephone Adapter will not re-establish the connection. This box is checked by default.
MTU	Enter the MTU (Maximum Transmission Unit) size. MTU is the largest size packet that can be sent by the device. If the network stack of any packet is larger than the MTU value, then the packet will be fragmented before the transmission. During the PPP negotiation, the peer of the PPP connection will indicate its MTU and will accept any value up to that size.

MAC Address Clone

Basically, every Ethernet interface (also called a NIC - Network Interface Card) has a MAC address. If the administrator assigns IP addresses to devices according to the authorized MAC address, or only allows devices with authorized MAC addresses to access network, you may manually enter the MAC address here.

Product Page: DVG-2101SF	>			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lini	K				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
Wizard	MAC ADDRESS C	LONE			Helpful Hints
Internet Setup VoIP Setup	Some cable Internet serv the PC that was original	vice providers require that dev y registered with the cable Int	rice connected to the cable mod ernet provider.	lem has MAC address of	Some cable Internet service providers require that device connected to the cable modem has MAC
Time and Date	MAC ADDRESS S	ETTINGS			address of the PC that was originally registered
Logout	M	MAC Address Clone	Cancel		with the cable Internet provider. More
BROADBAND					

MAC Address CloneCheck this checkbox to enable the MAC Address Clone feature.MAC AddressEnter the MAC address you want to use.

You can find the MAC address of your PC from **Start** > **Run**. In the run box type cmd and click **OK**.

For Windows 9x/ME, type **winipcfg** at the prompt and press **Enter**. The MAC address is listed as the "Adaptor Address".

For Windows 2000/XP, type ipconfig /all at the prompt and press Enter. The MAC address is listed as the "Physical Address".

Static Routing

You can specify additional information according to the network topology the VoIP Adapter is attached to in order to establish an efficient networking environment.

Product Page: DVG-2101SF	> Site Map	Firmware Version: 1.00.378
D-Lin1	K	
DVG-2101SP	SETUP ADVANCED MAINTENANCE STATUS	HELP
Wizard Internet Setup VoIP Setup	STATIC ROUTING LIST	Helpful Hints You can select what gateway to use, by interface or by specifying
Time and Date	Network Destination Netmask Gateway Enable	a gateway.
Logout	Add STATIC ROUTING	More
	Network Destination: Netmask: Gateway: 	
	Apply Cancel	
BROADBAND		

Click the **Add** button to configure Static Routing settings.

Network Destination	Enter in the IP address of the specified network that you want to access using the Static Route.
Netmask	Enter in the subnet mask for the specified network.
Gateway	Enter in the gateway IP address of the specified network.

VoIP Setup

The VoIP (Voice over Internet Protocol) Adapter can be configured to handle voice communication over the Internet. The screen shown below, along with the SIP account list and phone book pages are used to configure the basic settings to allow you to communicate with the devices that will send and receive telephone calls over the Internet.

SIP Account List

The VoIP Adapter allows you to make calls through the Internet or PSTN. You need to configure the SIP account before you can make a VoIP phone call.

D-Link strue strue strue strue strue strue strue strue strue strue strue strue strue strue strue strue strue strue strue strue<	Product Page: DVG-2	101SP			▶ <u>Site</u>	Map Firmware Version: 1.00.378
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Image: Second	VoIP Setup	Phone Port:	1 *			for making calls via Internet or PSTN. You can
Image: Sector Procession Rame: Image: Sector Procession Rame: Image: Research Pressword: Image: Research Image: Research Image: Research Play Name Enter your display name. Every outgoing call will carry this name to the receiver an should appear on the Caller-ID. Image: Research Image: Research r Name Enter your SIP account username. The username can be letters like in an e-mail address (e.g. account@provide com) on numbers like a telephone number (e.g. 24680@provider.com). Enter the authorization nam	Time and Date	Enable	_			assign SIP account the relevant phone number for incoming and outgoing
Authorization Name: Image: Constant on Name: Proxy Server: 0.0.0 Doubloand Proxy Server: 0.0.0 Due bornain to Register: 0.0.0 Expire Time: 0.0.0 Expire Time: 0.0.0 Proxy Server: 0.0.0 Expire Time: 0.0.0 Image: Time: 0.0.0 Expire Time: 0.0.0 Expire Time: 0.0.0 Expire Time: 0.0.0 Image: Time: 0.0.0 Enter your SIP account username. The username is the part that comes before the symbol. A username can be letters like in an e-mail address (e.g. account@provide com) or numbers like a telephone number (e.g. 24680@provider.com). horization Name Enter the authorization name to authenticate your SIP account. This is usually the st to your SIP account's user name. sword Enter the SIP proxy server's IP address and port number in this field (e.g. 192.168.0.253:5060). Domain to Register Required in order for some Australian ITSPs.	Logour	User Name:	le:	isername		calls.
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bitEnter the SIP proxy server's IP address and port number in this field (e.g. 192.168.0.253:5060).bound ProxyEnter the outbound proxy server IP address and port number in this field (e.g. 192.168.0.1:5060).e Domain to Register Required in order for some Australian ITSPs. Enter the amount of time for the dial tone when the phone has been lifted off the ho At the end of this time period, if no numbers have been dialed, the device will start playing the busy tone.VICheck the checkbox to activate MWI (Message Waiting Indication) if the SIP server supports this service	sword	Enter the passwo	ord to authen	ticate your SIP ad	count.	
 192.168.0.253:5060). Enter the outbound proxy server IP address and port number in this field (e.g. 192.168.0.1:5060). e Domain to Register Required in order for some Australian ITSPs. Dire Time Enter the amount of time for the dial tone when the phone has been lifted off the hore At the end of this time period, if no numbers have been dialed, the device will start playing the busy tone. VI Check the checkbox to activate MWI (Message Waiting Indication) if the SIP server supports this service 	xy/Backup Server	Enter the SIP pro	oxy server's l	P address and po	rt number ir	n this field (e.g.
 Enter the outbound proxy server IP address and port number in this field (e.g. 192.168.0.1:5060). E Domain to Register Required in order for some Australian ITSPs. Enter the amount of time for the dial tone when the phone has been lifted off the hore At the end of this time period, if no numbers have been dialed, the device will start playing the busy tone. VI Check the checkbox to activate MWI (Message Waiting Indication) if the SIP server supports this service 		192.168.0.253:5	060).			
 192.168.0.1:5060). e Domain to Register Required in order for some Australian ITSPs. bire Time Enter the amount of time for the dial tone when the phone has been lifted off the hore At the end of this time period, if no numbers have been dialed, the device will start playing the busy tone. VI Check the checkbox to activate MWI (Message Waiting Indication) if the SIP server supports this service 	ound Proxy	Enter the outbou	nd proxy ser	ver IP address an	d port num	per in this field (e.g.
 Domain to Register Required in order for some Australian HSPs. Enter the amount of time for the dial tone when the phone has been lifted off the hore At the end of this time period, if no numbers have been dialed, the device will start playing the busy tone. Check the checkbox to activate MWI (Message Waiting Indication) if the SIP server supports this service. 		192.168.0.1:506	U).			
At the end of this time period, if no numbers have been dialed, the device will start playing the busy tone. Check the checkbox to activate MWI (Message Waiting Indication) if the SIP server supports this service.	Domain to Kegis	ter Required in or	der for some	e Australian IISPs	, , the share	haa haan liftad aff tha ha
playing the busy tone. Check the checkbox to activate MWI (Message Waiting Indication) if the SIP server supports this service	re lime	Liller the amoun	time pariod	ine dial tone where h	i trie phone	rias been litted off the ho
Check the checkbox to activate MWI (Message Waiting Indication) if the SIP server supports this service		nlaving the bucy				ineu, ine uevice will staft
supports this service	1	Check the check	tone. hay ta setiva	te MWI (Meesaa	Waiting Ind	lication) if the SIP conver
	•	sunnorte this se	vice	to mining (mossayo	waiting inu	

Phone Book

This page allows you to add and edit your frequently used VoIP phone numbers in the phone book. It is also required if you want to make peer-to-peer calls. In peer-to-peer calls, you call another VoIP device directly without going through a SIP server.

Product Page: DVG-2101S	P Site Map	Firmware Version: 1.00.378
D-Lin l	K	
DVG-2101SP	SETUP ADVANCED MAINTENANCE STATUS	HELP
Wizard	PHONE BOOK LIST	Helpful Hints
Internet Setup		This page allows you to add and edit your
VoIP Setup	Phone Book: Page 1 V	frequently used VoIP phone number in the
Time and Date	No. Phone Number Display Name SIP URL	phone book.it is also required if you want to make peer-to-peer calls. In
		peer-to peer calls, you call another VoIP device
		directly without going through a SIP server.
	Add	
	РНОЛЕ ВООК	More
	Phone Number:	
	Display Name:	
	Distinctive Ring: Group0 👻	
	SIP URL: @	
	Apply Cancel	
BROADBAND		

Click the **Add** button to edit the phone book.

Phone Number Display Name	Enter the phone number of your contact. Enter a name to display on telephone set's LCD screen when there is an incoming call from your contact.
Distinctive Ring	Select a distinctive ring type from the drop-down list to classify the incoming calls into different categories such as family, friends, or colleagues.
SIP URL	Enter the SIP URL of your contact. The format of a SIP URL is similar to an e-mail account. The format should be ID@itsp.com.

Time and Date

The VoIP Adapter allows you to maintain the system clock via the Internet. You must first select your time zone from the drop-down list, then select a pre-defined NTP (Network Time Protocol) time server from the drop-down list, or specify the IP address of your preferred NTP time server on the Internet manually.

Synchronizing with a time server ensures the time-based client filtering features and system log entries are based on the correct localised time.

Product Page: DVG-2101S	P			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lini	K				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
Wizard	TIME AND DATE				Helpful Hints
Internet Setup VoIP Setup Time and Date	The Time Configuration option a system clock. From this section Protocol) Server. Daylight Savir	allows you to configure, you can set the time zor ng can also be configured	update, and maintain the corre le that you are in and set the i d to automatically adjust the ti	ect time on the internal NTP (Network Time me when needed.	If you plan on using the scheduling feature of this VoIP TA, then making sure the time is correct is extremely important. Fither enter the time
Logout	NTP SETTINGS				manually by dicking the Copy Your Computers
	 Automatically sync First NTP Time Server: Second NTP Time Serve 	hronize with Internel	t •		Time Settings Dutton, or use the Automatic Time Configuration option to have your VOIP TA synchronize with a time server on the Internet.
	TIME CONFIGURATION				More
	Current Device Time:	Fri May 16 12:45:0	9 2008		
	Time Zone:	(GMT-7:00)Mountai	n Time(USA&Canada)	•	
	Enable Daylight Sav	ving			
	Daylight Saving Offset:	-2:00 🔻	h Davi 70		
	Daylight Saving Dates:	Start Jan v 1st	x Day Time v Sun 12 am v v Sun 12 am v		
		Apply	Cancel		
BBOODBODD					

Automatically Synchronize with Internet	Check this checkbox and then select the preferred time server(s) from the drop- down NTP Time Server list.
Time Zone	Select the time zone where you reside from the drop-down list.
Enable Daylight Saving	Check this checkbox if your location uses daylight saving. Then select the daylight saving offset from the drop-down list.
Daylight Saving Dates	Select the start and end date for daylight saving to take effect from the drop- down list.

Advanced

VoIP Settings

This VoIP section allows you to configure advanced settings for your VoIP connections.

Call Feature

This page allows you to configure advanced features for your SIP account.

This page anows yo	Su to configure a			ount.			
Product Page: DVG-2101SP						▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Link	Ć						
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE		ST	ATUS	HELP
VoIP Settings							Helpful Hints
DNS	CALL FEATURE						Call Feature Services such
Dynamic DNS		Phone Port: 1 v					as consultation hold, call waiting, call back on busy, are generally available
QoS		Forward Always:					provider.
VPN		Forward Busy:					
Logout		Forward No Answer:		Wait:	15	s(1-20)	More
		Warm Line:		Wait:	30	s(1-30)	
		Hot Line:					
		Do Not Disturb					
		Call Waiting					
		Consultation Hold					
		Conference					
		Call Back On Busy					
		Apply	Cancel				
BROADBAND							
Forward Always	Check the ch	eckbox and enter t	he phone numb	er to	forwar	d all incor	ning calls to,
	regardless o	f any other conditio	ns.		,		
Forward Busy	Check the ch	eckbox and enter t	ne pnone numb	er to	torwar	d all incor	ning calls to, if the
Forward No Answe	Check the ch	o DUSY. Jeckhox and enter t	he nhone numh	er to	forwar	d all incor	ning calls to if the
rorward no Answe	call is not an	swered. You must	also enter the n	umbe	er (in se	econds) th	nat the VoIP
	Adapter sho	uld wait for you to a	answer an incor	ning	call bef	ore it con	siders the call
	unanswered	, ,		Ū			
Warm Line*	Check the ch	eckbox and enter a	specific numb	er to I	be diale	ed automa	tically by the VoIP
	Adapter whe	n the user goes off	-hook on the lir	ie and	does i	not input a	any digits after a
	configurable	timeout has elapse	d. For example	, this	teature	can be u	sed for emergency
Hot Line*	Or operator (naiing it you do not	anput any diali	ng ini or to l	ormati	on atter a	certain time.
	by the VolP	danter when the u	ser ages off-bo	unk or	ue utale n the lin	e nynt av	vay automatically
		auptor whom the u					

Do Not Disturb	Check this checkbox to reject incoming calls as busy or divert incoming calls to a voice mail server when configured to do so.
Call Waiting	Check the checkbox to receive a second call when a call is already ongoing on the same phone port.
Consultation Hold	Check the checkbox to let you to place a call on hold, and then call another party to consult privately, and afterward, return to the original call.
Conference	Check the checkbox to enable making conference calls.
Call Back On Busy	Check the checkbox to enable call back on busy when the number you dialed is busy.
PSTN Fail Over Relay	Check the checkbox to enable PSTN fail over relay when the Internet connection is not available.

*You can only activate either Warm Line or Hot Line. If you check both of them, Hot Line's priority is always set to a higher priority than to that of Warm Line.

Examples of Making a Conference Call

Scenario 1:

- 1. The first person dials the second person and asks the second person to wait.
- 2. The first person presses Flash then calls the third person (and keeps the second person on hold).
- 3. The third person answers the call. The first person dials *61 and then presses Flash to start the conference call.

Scenario 2:

- 1. The first person dials the second person and the second person answers the call.
- 2. The third person dials the first person (call waiting) and the first person presses Flash to pick up the second call and talk to the third person.
- 3. The first person dials *61 and then presses Flash to start the conference call.

Note: The availability of conference calls varies depending on your VoIP network. Please check with your ITSP for this service.

SIP Setting

This page allows you to configure the settings related to the SIP service provider. You need to configure your VoIP Adapter to communicate with the devices that will send and receive telephone calls over the Internet. If you are behind a NAT modem or router, you may have to use a STUN server.

Product Page: DVG-2101S	P			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lin l	k				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
VoIP Settings	SIP SETTING				Helpful Hints
DNS	SIF SETTING				This page allows you to set
Dynamic DNS	Port				to the SIP service
QoS		Signaling Port:	5060		configure your VoIP device
VPN		RTP Start:	40000		devices that will send and
Logout	NAT Setting	RTP End:	50000		over the Internet. If you
	intr Second	STUN			are behind a NAT modem/device, you may
		Server Address:	larry.gloo.net		have to use a STUN server.
		NAT Keep Alive Time:	60 sec (30≤X:	≤86400)	
		Rport			More
	T.38 Fax				
	Advanced Feat	T.38 FAX Protocol			
		P_Assert			
		E.164 ENUM			
		Session Timer:	1800 sec (90≤X))	
		Anonymous Call Rejection	on		
		DTMF INFO Type For Cisc	0		
		Prack			
		Apply	Cancel		
BROADBAND					

Signaling Port

	The default value is 5060 and this must match with the peer Registrar when making VoIP calls.
RTP Start	Specifies the start port for RTP stream. The default value is 40000.
RTP End	Specifies the end port for RTP stream. The default value is 50000.
STUN	Enable or disable STUN server feature.
Server Address	Enter the STUN server IP address if you enabled the STUN server feature.
NAT Keep Alive Time	Check the checkbox and enter the time period for the NAT to keep alive.
Rport	Check this checkbox to use rport. Rport, or symmetric signaling, is critical for operations behind some classifications of NATs. If enabled, the SIP stack will send outbound SIP messages on the same port that it listens on. Unless you are familiar with single SIP port, it is better to use rport.
T.38 Fax Protocol	Check the checkbox if the remote end also supports FAX function.
P_Assert	Check the checkbox to enable P_Assert to hide the client name when sending messages to the SIP proxy server.

Enter the signaling port to send and receive SIP message for building a session.

E.164 ENUM	Check the checkbox to enable E.164 ENUM (Number Mapping) feature. E.164 is a mechanism to transform an E.164 number into a list of URIs (Uniform Resource Identifiers) so that the IP users and available services can be identified by a public telephone number.
Session Timer	Check the checkbox to enable this function. If there is no response, the connecting call will be closed automatically after the time limit that you set in this field.
Anonymous Call Rejection	Check the checkbox to reject the all anonymous calls.
DTMF Info Type	Check this checkbox to use DTMF (Dual Tone Multi-Frequency) to send out the Caller ID.
Prack	Provides reliability of Provisional Responses in the Session Initiation Protocol (SIP)

Digit Plan

This page enables you to set up rules for the digit waiting time and the digit plan. A digit plan is also known as a dial plan. A digit plan is a string of characters that governs the way the VoIP Adapter processes inputs received from the telephone keypad. This includes country codes, access codes, area codes, and all combinations of digits dialed. Digit plans must comply with the telephone networks to which they connect.

Product Page: DVG-2101SP				▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Link	¢				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
VoIP Settings	DIGIT PLAN				Helpful Hints
DNS Dynamic DNS QoS VPN Logout	First Digit Waiting T Interval Digit Waiti Digit Plan:	Fime: 12 ing Time: 10	s (1-20) s (1-20)		This page enables you to set up rules. More
80008000					
BROADBAND					

First Digit Waiting Time	Enter the first digit waiting time before the connection is canceled by the VoIP Adapter.
Interval Digit Waiting Time	Enter the interval waiting time between two digits before the connection is canceled by the VoIP Adapter.
Digit Plan	Enter the strings for the digit plan.

Digit Plan Syntax

To specify a	Enter the following	Result
Digit	0123456789*	Identifies a specific digit (do not use #)
Range	[digit-digit]	Identifies any digit dialed that is included in the range
Range	[digit-digit, digit]	Specifies a range as a comma separated list
Wildcard	X	x matches any single digit that is dialed
Wildcard		. matches an arbitrary number of digits
Time	Т	Indicates that an additional time out period of 4 seconds should take place before automatic dialing starts

For example, let's look at the following digit plan:

<:1780>[2-4]xxxxxx|*xx|<911:17804213333>|011[2-9]x.T|1[2-9]xx[2-9]xxxxxx|1900xxxxxxx!

A digit plan is the combination of one or more dialing rules. The above digit plan includes six dialing rules and is separated by five "|" characters. The VoIP Adapter interprets the digit plan (strings) in order from left to right. That is, if you dial a set of numbers that fits to more than one section of the digit plan, the VoIP Adapter will use the first numerical sequence that fits to process your inputs.

Component	Description
<>	A pair of <> brackets is used to enclose digit and character for special usage such as add prefix to the following digit(s) or replace the digits you dialed. For example, <:111>*123 means that when you press 123 on the keypad, the VoIP Adapter will send out 111*123. <*123:#911> means when you dial *123, the VoIP Adapter will send out #911.
:	The ":" means replace any digits within the <> brackets before the ":" with the digits written after the ":"
	For example: <:1780> means the system prefixes all numbers that match with the remainder of the dial plan string with "1780". <911:17804213333> means the system will send out the call with 17804213333 whenever you dial 911. If you live in Australia and your VoIP provider resides in North America, your may change this dial plan to <000:011612131444>.
[]	A pair of [] brackets is used to enclose one digit (the digits on telephone keypad *, # and 0-9). For example, [*#2-9] means the digit can be either *, #, 2, 3, 4, 5, 6, 7, 8 or 9. [*1-3579] mean the digit can be either *, 1, 2, 3, 5, 7 or 9.
Х	The x represents any digit from 0 to 9. x means one 0-9 digit, xx means two 0-9 digits and xxx means three 0-9 digits, etc.
	For example, [2-4]xxxxx means you are allow to dial either 2,3 or 4 followed by any six digits from 0 to 9.
 *	The " " in a digit plan is merely used to separate the different dialing rules. The "*" represents the * (star key) on your telephone keypad. For example, *xx means press * key followed by any two digits from 0-9, such as *69 or *55.
	The "." (dot) means repeat the previously listed digit zero or more times. For example, x. means zero or more digits and xx. means at least one or more digits.
Т	The "T" represents that an additional timeout period should take place before automatic send out the dialing.
!	The "!" in a digit plan means the digit or number is restricted. For example, 11[0, 2, 4, 9]! means the system will block the numbers 110, 112, 114 and 119. 00886! means that the dial out number beginning with 00886 is restricted. Note that the "!" can only be placed at the end of a dialing rule.

The VoIP Adapter processes call features and the digit plan in the following sequence:



* Including keypad dialing, hotline, warm line, etc.

If you configure the dialing rules or behaviors in the VoIP Adapter as: Rule 1: in dial prefix settings, set *666 for IVR accessing Rule 2: in phone book settings, set *666 to map to the URL: abc@domain.com Rule 3: in the digit map, set *666 to be replaced by 911 Rule 4: in the digit plan, set *666 to be accepted and 911 to be restricted When there are rules 1 to 4 in the system, pressing *666 will enter the VoIP Adapter's IVR mode. When there are rules 2 to 4 in the system, pressing *666 will make a VoIP call to abc@domain.com When rule 3 and rule 4 are in the system, pressing *666 will be replaced with 911 by the VoIP Adapter, but this number will not be sent out.

When there is only rule 4 in the system, pressing *666 will immediately send the dialing to the proxy or UA.

When you dial	*666:	
Scenario	Rule Combination	The VoIP Adapter will do
1	Rule1 + Rule2 + Rule3 + Rule4 =	Enter the VoIP Adapter's IVR mode
2	Rule2 + Rule3 + Rule4 =	Address a call to abc@domain.com
3	Rule3 + Rule4 =	Replace the input with 911, but this number will not be sent out
4	Rule4 =	Immediately send the dialing to the proxy or UA

If the dialed numbers or characters do not match to the digit plan, you will probably hear a busy tone from the telephone handset.

Digit Map

This page enables you to set up rules for modifying dialed numbers, or varying the dialing behavior depending on the number called. The digit map specifies how to interpret digit sequences dialed by the user, and how to convert those sequences into an outbound dial string.

Product Page: DVG-2101S	Ρ				▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lin l	k					
DVG-2101SP	SETUP	ADVANCED	MAINTENAN	ICE	STATUS	HELP
VoIP Settings	DIGIT MAP LIST					Helpful Hints
DNS Dynamic DNS	Digit Map: 🔻					This page enables you to set up rules for modifying dialed numbers, or varying the dialing behavior
QoS VPN	No. Prefix	Mode OP Nur	mber Min Len	Max Len	Interface	depending on the number called.
Logout						More
			Add			noreal
	DIGIT MAP					
		Prefix:				
		Mode:	None 🔻	-		
		OP Number:				
		Max Len:				
		Interface:	Phone 1 👻			
		Apply	Cancel			
BROADBAND						

Click the **Add** button to add a digit map.

Prefix	The prefix specifies a series of digit sequences. When you dial a series of digits, each sequence in the prefix is tested as a possible match. The matching sequences form a set of candidate digit sequences. The phone number set here is used to add, strip, or replace the OP number.
Mode	None: No action.
	Add: When you choose this mode, the OP number will be added with the prefix number for calling out through the specific phone interface.
	Replace: If you select this mode, the OP number will be replaced by the prefix number for calling out through the specific phone interface.
	Delete: If you select this mode, the OP number will be deleted by the prefix number for calling out through the specific phone interface.
	Restrict: If you select this mode, the OP number will be restricted by the prefix number for calling out through the specific phone interface.
OP Number	The front number you type here is the first part of the account number that you want to execute a special function (according to the chosen mode) by using the prefix (number). If you selected the mode "Add", "Replace", or "Delete", the OP number will be applied in this rule.
Min Len	Set the minimal length of the dial number for applying the prefix number settings.

Max LenSet the maximum length of the dial number for applying the prefix number settings.InterfaceYou can apply the prefix number settings to the PHONE port (Phone 1).

Diait	Manu Dage1	~						
Digit	Map. [Pager							
No.	Prefix	Mode	OP Number	Min Len	Max Len	Interface		
1	03	Add	886	6	8	Phone 1	ſ	
2	06	None		6	8	Phone 1	ſ	
3	9	Replace	0800	6	8	Phone 1	F	1

- The first rule accepts calls using 03 as the prefix number. For example, if you dial "038993765", it matches this rule and the VoIP Adapter will make the phone call "886038993765".
- The second rule accepts calls using 06 as the prefix number. For example, if you dial "065991234", it matches this rule, and its length (not including the length of the prefix) is 7, between 6 (Min Len) and 8 (Max Len), therefore the VoIP Adapter will make the phone call "065991234".
- The third rule accepts calls using 9 as the prefix number. For example, if you dial "93652400", it matches the third rule and the VoIP Adapter will make a phone call "08003652400".

Dial Prefix

This page allows you to configure your dialing policy. You can change the default dial prefix settings to fit your dialing behaviour.

Product Page: DVG-2101S	SP.				▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lin	K					
DVG-2101SP	SETUP	ADVANCED	MAINT	ENANCE	STATUS	HELP
VoIP Settings	DIAL DREETY SETT	ING				Helpful Hints
DNS	DIAL PREFIX SET	ING				This page allows you to
Dynamic DNS	🗹 Dial P	refix				policy.
QoS	System R	eboot:	*444	(Default:*4	44)	
VPN	Factory D	efault:	*555	(Default:*5	55)	More
Logout	Net Settin	ng:	*666	(Default:*6	66)	
	Call Retur	n:	*50	(Default:*5	0)	
	Repeat Di	ialing:	*52	(Default:*5	2)	
	Fail Over	Relay:	*54	(Default:*5	4)	
	PSTN Fail	Over Relay Active:	*55	(default:*5	5)	
	Anonymo	us Call Rejection Active:	*56	(Default:*5	6)	
	Anonymo	us Call Rejection Deactive:	*57	(Default:*5	7)	
	Caller ID I	Block:	*62	(Default:*6	2)	
	Hot Line A	Active:	*64	(Default:*6	4)	
	Do Not Dis	sturb Active Code:	*66	(Default:*6	6)	
	Do Not Dis	sturb Deactive Code:	*68	(Default:*6	8)	
	Forward A	Always Active Code:	*72	(Default:*7	2)	
	Forward A	Always Deactive Code:	*73	(Default:*7	3)	
	Forward B	Busy Active Code:	*90	(Default:*9	0)	
	Forward B	Busy Deactive Code:	*91	(Default:*9	1)	
	Forward N	loanswer Active Code:	*92	(Default:*9	2)	
	Forward N	loanswer Deactive Code:	*93	(Default:*9	3)	
	Call Back	On Busy Cancel:	*61	(Default:*6	1)	
	Call Back	On Busy Auto Enable:	*58	(Default:*5	8)	
	Call Back	On Busy Auto Disable:	*59	(Default:*5	9)	
		Apply	Cancel			

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Codec

The codec used for each call will be negotiated with the peer party before each session, and therefore many codecs will not be included in your default configuration. The default codec is G.729A, it occupies little bandwidth while maintaining good voice quality.

If your upstream speed is only 64Kbps, using G.711_A codec is not recommended. It is better to have at least 256Kbps upstream if you would like to use G.711_A.



```
Packetization TimeSelect the length of the digital voice segment that each packet holds. The default is 20<br/>millisecond packets. Selecting 10 millisecond packets enhances the voice quality, as<br/>less information is lost due to packet loss, but doubles the load on the network traffic.DTMFSelect the method of DTMF tone generation from the drop-down list.
```

Phone Setting

This page allows you to enable VAD (Voice Activity Detection), CNG (Comfortable Noise Generation), and Echo Cancellation. You can also specify the Receive/Send Gain as well as select the Telephony Tone and Caller ID type. The VoIP Telephone Adapter provides default ring-tone parameters configured for various countries.



Voice Activity Detection	The VAD (Voice Activity Detection) feature allows the VoIP Adapter to stop transmitting when you are not speaking. This reduces the bandwidth the VoIP Adapter uses.
Comfortable Noise	The CNG (Comfortable Noise Generation) feature allows the VoIP Adapter to
Generation	generate comfortable background noise to indicate that you are still on the line.
Echo Cancellation	Enable the Echo Cancellation feature to eliminate the echo caused by the sound of your voice reverberating in the telephone transmitter while you talk.
Receive Gain	Enter the receive gain value.
Send Gain	Enter the send gain value.
Telephony Tone	Select the telephony tone type from the drop-down list according to your region.
Caller ID	Select the caller ID type from the drop-down list according to your region.

DNS

The DNS (Domain Name System) feature allows you to set the configuration of the DNS proxy.

Product Page: DVG-2101S	P			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Linl	k				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
VoIP Settings DNS Dynamic DNS QoS VPN Logout	DNS server is used for tran DNS SERVER CONF Obtain D Use the f Preferred Secondary	nslating a URL to an IP addr FIGURATION NS server address auton following DNS server add DNS Server : DNS Server : Apply	ess.	(Optional) (Optional)	Helpful Hints If Use the following DNS server addresses is selected, enter the Preferred and optional Alternate DNS server IP addresses. More
BROADBAND					

DNS Server Configuration

Primary DNS Address Secondary DNS Address Select to obtain DNS server address automatically or specify the DNS server(s) if you are provided DNS information by your ISP. Enter the primary DNS IP address provided by your ISP. Enter the secondary DNS IP address provided by your ISP. This is an optional DNS Address entry to be used if the primary DNS server fails.

Dynamic DNS

The Dynamic DNS service allows you to bind the device's public IP address to a static hostname, allowing your computer to be more easily accessed from various locations on the Internet. Most broadband Internet Service Providers assign dynamic (changing) IP addresses. Using a DDNS service provider, your friends can enter in your domain name to connect to your server no matter what your IP address is. When this feature is enabled, this device will automatically update your DDNS records whenever your IP address changes.

Product Page: DVG-2101SF	>				Þ	<u>Site Map</u>	Firmware Version: 1.00.378
D-Lin1	K						
DVG-2101SP	SETUP	ADVANCED	MAINTENA	NCE	STATU	IS	HELP
VoIP Settings	DYNAMIC DNS						Helpful Hints
DNS Dynamic DNS QoS	The DDNS feature allows you to have purchased(www.whateve Internet Service Providers assig can enter your host name to co	nat you dband friends	DDNS - This stands for Dynamic DNS. Allows you to bind the device's public IP address to a static hostname				
VPN	Sign up for D-Link's Free DDNS	service at <u>www.DLinkD[</u>	DNS.com.				users will be able to point to this in order to access a
Logout							dynamic IP address from anywhere in the world.
	DDNS SETTINGS						
	Enable Dynamic DM	IS					More
	Server Address:	www.DLinkDDNS.co	m <<	www.DLir	nkDDNS.com	Ŧ	
	Host Name:						
	Username or Key:						
	Password or Key:						
	Verify Password or Key	/:					
	Timeout:	1	(hours)				
	Status:	Disconnect					
	L	Apply	Cancel				
BROADBAND							

Server Address	Enter your DDNS server's IP address provided by your DDNS service provider. You can also select a pre-defined DDNS service provider from the drop-down list.
Host Name	Enter the host/domain name that you registered with your DDNS service provider.
Username or Key	Enter the username or key for your DDNS account.
Password or Key	Enter the password or key for your DDNS account.
Verify Password or Key	Re-enter the password or key for your DDNS account.
Timeout	Enter the DDNS timeout period in hour(s).
Status	This field shows the DDNS connection status as connected or disconnected.

QoS

This QoS (Quality of Service) feature allows you to classify traffic by its priority and mark it with a Differentiated Services (DiffServ) flag.

Product Page: DVG-2101S	P			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lini	k				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
VoIP Settings	QUALITY OF SERV	ICE			Helpful Hints
DNS Dynamic DNS QoS	The bandwidth gap betwe applications, such as VoIF mark it with differentiated	een LAN and WAN may signifi ?, gaming, and VPN. This QoS I services (Diffserv) flag.	cantly degrade performance of function allows users to classif	critical network y traffic by its priority and	This QoS function allows you to dassify traffic by its priority and make it with differentiated services (DiffServ) flag.
VPN	SETTINGS				
Logout					More
		Qo5 802.1Q VLAN T	AG Enable		
		VLAN ID:	(0-4	094)	
		Priority Voice :	(0-7)	
		Priority Data :	(0-7)	
		Apply	Cancel		
BROADBAND					

VLAN ID

Specify the VLAN ID from 0 to 4095.

Priority (Voice/Data) Indicate the priority of voice and data separately. The higher the value, the higher the priority. If you configure data and voice with the same value, it means they have the same priority.

> You can set the voice priority to the highest value 7 in order to ensure the best voice quality.

VPN

A Virtual Private Network (VPN) is the extension of a private network that encompasses links across shared or public networks like the Internet. In short, using VPN technology, you can send data between two computers across a shared or public network in a manner that emulates the properties of a point-to-point private link.

Dialing out Settings

This page allows you to configure the dialing out setting to be IPSec (Internet Protocol Security), PPTP (Point-to-Point Tunneling Protocol), or L2TP (Layer 2 Tunneling Protocol) and its local IP and remote IP(s).

Click the **Add** button to create a profile for the VPN connection.

Product Page: DVG-2101SP	•			Site Map	Firmware Version: 1.00.378
D-Lini	¢				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
VoIP Settings DNS Dynamic DNS QoS VPN Logout	ACCOUNT 1	Enable count Name: ype: PPTP - ser Name: assword: erver IP: Auto Reconnect Use Remote VPN Gat Apply	eway As The Default Gatev	vay	Helpful Hints The Dialing Out settings has four groups can setup. Setup the tunnel through PPTP/LZTP Server and PPTP/LZTP Client according to different forms. More
BROODBODD					
DROADDAND					
Account Name	Enter an acco this device.	unt name for the o	connection. You can	configure up to f	our accounts for
Type User Name Password Server IP	Select the VPI Enter the user Enter the pass Enter the serv VPN gateway	N type to be IPSec r name for VPN co sword for VPN con rer IP for VPN con as the default gat	c, PPTP, or L2TP fro onnection. nnection. nection and/or chec eway.	m the drop-down k the checkbox to	list. use the remote
Kemote LAN IP	Enter the rem	ote LAN IP for the	VPIN connection.		

Remote LAN Mask Enter the remote LAN mask for the VPN connection.

Maintenance

System

Configuration

These configuration tools help you reboot, backup, restore settings, or reset the VoIP Adapter to its factory default settings.

Product Page: DVG-2101S	Р			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lin l	K				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
System	CONFIGURATION 1	TOOLS			Helpful Hints
Firmware Update Access Control Ping Test	Use the "Backup" tool to s use the "Restore" tool to the "Restore to Factory D the original factory setting	save the Telephone Adapter's restore the saved configuratio efaults" tool to force the Tele gs.	current configuration to a file in to the Telephone Adapter. Shone Adapter to perform a p	on your PC. You can then Alternatively, you can use sower reset and restore	This page allows you to reboot your VoIP TA or save your VoIP TA configuration to a file on your computer as a
Logout	SYSTEM REBOO Click the reboot button to	precaution in case you have to reset your VoIP TA to factory default settings. You will be able to restore your VoIP TA settings from a previously saved configuration file. There is also a function to allow you to reset your			
	SYSTEM BACKU Backup VoIP TA configura	P SETTINGS tions. You may save your VoIA Backup	• TA configurations to a file or Settings	n your PC.	VoIP TA to factory default settings. Resetting your VoIP TA to factory default settings will erase your current configuration.
	SYSTEM UPDAT	E SETTINGS			More
	Update VoIP TA settings.	You may update your VoIP TA	settings using your saved file	es.	
	Settir	ngs File Name:	Brow	wse	
	SYSTEM RESTO	RE DEFAULT SETTING	s		
	Restore VoIP TA settings	to the factory defaults.	ault Settings		
BROADBAND					

Reboot

This feature allows you to reboot the device while still keeping its current settings.

Backup SettingsThis feature allows you to save the device's current settings as a file on your PC.Update SettingsThis feature allows you to update the device's current configuration with your
previously saved files.

Restore Default Settings This feature allows you to force the device to perform a power reset and restore the original factory default settings.

Auto Provision

This page allows you to automatically update your VoIP Adapter's configuration by using a TFTP, FTP, HTTP, or HTTPS server located on the Internet. If you have to access such services, you will need to know the IP address of the provisioning server.

Product Page: DVG-2101S	2) <u>Site Map</u>	Firmware Version: 1.00.378
D-Lin l	K				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
System Firmware Update Access Control Ping Test Logout	AUTO PROVISION Au Pro Sec Po Pro File	to Provision: Only ptocol: TFTP rver: 69 ovision Times: 0 riod: 1 e: back0.ba	Power Up O Interval O (Optional) Days O Hours O k Cancel	Disable Minutes	Helpful Hints Provisioning is a function that automatically updates your VoIP device's configuration by using a TFTP, FTP, HTTPS or HTTP server located on the Internet. If you have access to such service, you will need to know the URL or IP address of the Provisioning Server. Note: Fill in the parameters needed by your VoIP Service Provider. Please check with your VoIP Service Provider about the availability of these services. More
BROADBAND					
Auto Provision Protocol IP Port	Select you p Indicate the Enter the IP Enter the point number is 69	rovision criteria to provision protocol address supplied l rt number supplied 9.	be Only Power Up, supplied by your IS by your ISP for the d by your ISP for th	Interval, or Disable SP from the drop-de device to provision. e device to provisio	e. own list. n. The default port

Provision TimesEnter the number of times to provision.PeriodEnter the period to provision if you set your provision criteria as Interval.FileEnter the file name supplied by your ISP for the device to provision.

SNMP

SNMP (Simple Network Management Protocol) is a protocol allowing network administrators to obtain information and even configure various network devices remotely. A computer attached to the network is called a Network Management Station (NMS). The access rights to the agent are controlled by community strings. To communicate with the VoIP Adapter, the NMS must first submit a valid community string for authentication. This page allows you to configure basic access control, using passwords called community (strings).

Product Page: DVG-2101SP				▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Link	Ć				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
System	SNMP				Helpful Hints
Firmware Update Access Control Ping Test	SNI Rea	MP Version: 1 - ad Community: public			The Simple Network Management Protocol (SNMP) from part of the internet protocol suite ad defined by the Internet
Logout	Wr	ite Community: private			Engineering Task Force (IETF). SNMP is used by
	Tra	p Receive Server:	<u>. </u>		network management systems to monitor petwork-attached devices
		Apply	Cancel		for conditions that warrant administrative attention. It consists of a set of standards for network management, including an Application Layer protocol, a database schema, and a set of date objects. More
BROADBAND					
SNMP Version	Select the SN and v2 proto	IMP version from t cols.	he drop-down list.	The VoIP Adapter s	supports both v1
Read Community	Enter the rea public.	d community string	g for authorizing re	ad-only rights. The	e default string is
Write Community	Enter the wri string is priv	te community strin ate.	g for authorizing re	ead-and-write right	s. The default

Trap Receive Server Enter the trap receive server's IP address for the traps to be sent to this address when errors or specific events occur on the network.

Trap Community Enter the trap community string for authorizing trap management rights.

Log Setting This page allows you to configure settings to upload logs to a TFTP server and/or download logs to your local drive.

Product Page: DVG-2101S	P			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lini	k				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
System Firmware Update Access Control Ping Test Logout	UPLOAD SETTINGS	S Upload Enable Server IP: Ate Interval: File Size: 100	Days KB(1KB~200KB)		Helpful Hints Use the Log Settings screen to configure to where the VoIP Device is to upload logs; the schedule for when the VoIP Device is to send the logs and which logs and/or immediate the VoIP device is to send. More
BROADBAND					

TFTP Server IP	Enter the IP address of a TFTP server.
Update Interval	Enter the update interval in day(s).
Max File Size	Enter the maximum file size that can be accepted.
Log File Type	Select the log file type, either System Log or Network Statistics, from drop-down list to download logs.

Firmware Update

This tool allows you to upgrade this device's firmware locally. This device is also capable of being managed remotely by the service provider.

Note: Please contact your Internet Service Provider prior to upgrading your firmware. The update process takes several minutes to complete, and your VoIP Adapter will reboot. Please DO NOT power off your device before the update is complete.



Firmware File Name

Enter the path and name, or click the Browse button to locate the upgrade file then click the Update Firmware button. You will be prompted to confirm the upgrade to complete the process.

Access Control

Access to your VoIP Adapter is controlled through two user accounts: admin and user. The username "admin" has unrestricted access to change and view the settings of the VoIP Adapter. The username "user" can access the VoIP Adapter, view configuration settings and statistics, and update the VoIP Adapter's software. By default, all passwords are intentionally left as blank. You can change the administrator and common user's passwords respectively in this page. The VoIP Adapter will log you out automatically after the web interface is idle for a period of time. The default idle timeout value is 300 seconds. You can change the time period if necessary.

Product Page: DVG-21015	SP Site Map	Firmware Version: 1.00.378
D-Lin	K	
DVG-2101SP	SETUP ADVANCED MAINTENANCE STATUS	HELP
System Firmware Update Access Control Ping Test Logout	ADMIN PASSWORD Current Password: New Password: Confirm Password: Session Time: 0 seconds (60-32767, default:300, 0:never)	Helpful Hints For security reasons, it is recommended that you change the Login Name and Password for the Administrator and User accounts. Be sure to write down the new Login Names and Passwords to avoid having to reset the device in the event that they are forgotten.
	USER PASSWORD	
	New Password : Confirm Password :	More
	Apply Cancel	
BROADBAND		

Admin Current Password	Enter the administrator's current login password.
Admin New Password	Enter a new password for the administrator to login.
Confirm Password	Re-enter the new password for verification.
Session Time	Enter the session timeout duration (in seconds). The system will disconnect the access rights of the administrator when the session time is up.
User New Password	Enter a new password for the user to login.
Confirm Password	Re-enter the new password for verification.

Ping Test

The VoIP Adapter allows you to perform a ping test. This feature helps you to diagnose connection problems. If you can ping an IP on the WAN side successfully, you should be able to access the Internet.

Product Page: DVG-2101SF	Ρ			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lin1	K				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
System Firmware Update Access Control	PING TEST IP/D	omain :	Ping]	Helpful Hints These tools can be used to verify physical connectivity on both the LAN and WAN interfaces.
Ping Test Logout	The result of Ping			~	More
BROADBAND					

IP/Domain Enter the IP/Domain you want to ping in the field below and then click the Ping button.

Status

Device Information

You can use the Device Info screen to view the connection status for the VoIP Adapter's WAN/LAN interfaces, current firmware, and other related information.

Product Page: DVG-2101SF	0			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D.I imi	_°				
	~				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
Device Info	DEVICE INFORMA	TION			Helpful Hints
VoIP Status CDR VPN Status	You can use the Device In interfaces,firmware and I connected to your netwo	nformation screen to see the co hardware version numbers,as w rk.	onnection status for this devic vell as information on all DHCF	ce's WAN/LAN P dient PCs currently	All of your Internet and network connection details are displayed on the Device Info page. The firmware version is also
NetWork Statistic	SYSTEM INFO				displayed here.
System Log	Tananananan				
Logout	Module Name	DVG-2101SP			More
	Serial Number	S072300105			
	WAN MAC	00:50:56:C0:00:08			
	Hardware Version	0.4			
	Kernel Version	1.00.378			
	WAN				
	Ethernet	Connected			
	WAN IP	192.168.40.119(dhq)		
	Subnet Mask	255.255.255.0			
	Gateway	192.168.40.254			
	Primary DNS	192.168.0.249			
	Secondary DNS	192.168.0.243			
		Release	Renew		
		Refr	esh		
BROADBAND					

VoIP Status

You can use the VoIP Status screen to view the phone port status.

n: 1.00.378
P
⊷ ≥ VoIP
o see oort status.

CDR Status

You can use the CDR (Call Detail Record) Status screen to view the call history records of incoming and outgoing calls. It stores up to 100 records.

Product Page: DVG-2101S	P					Site Map	Firmware Version: 1.00.378
D-Lin l	k						
DVG-2101SP	SETUP	ADVANCED	MAINT	ENANCE	STA	TUS	HELP
Device Info VoIP Status CDR VPN Status NetWork Statistic	CDR LIST Port: 1 • First Page Page 1 of 1	Last Page Previo	us Next				Helpful Hints Call Detail Record feature provides history record of incoming and outgoing calls. It stores up to 100 records.
System Log	No. Display Name	Phone Direction	Proxy/Remote Host	Remote Phone No.	Start Time	Duration	More
BROADBAND							

VPN Status

You can use the VPN Status screen to view the VPN connection status.

Product Page: DVG-2101	SP			▶ <u>Site Map</u>	Firmware Version: 1.00.378
D-Lin	k				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
Device Info	VPN MANAGEMEN	г			Helpful Hints
VoIP Status	The VPN management scr	een allows you to check the V	PN status and to dial or to dro	p VPN connections.	Can see the situation of Dial Out through VPN
CDR.	DIAL OUT				Management.
NetWork Statistic					More
System Log	Name Type C	lient IP Server IP	UpTime Sta	atus	
Logout					
BROADBAND					

Network Status

You can use the Network Status screen to view the routing table, VoIP and network statistics to observe the current network conditions.

Product Page: DVG-2101SF	P					▶ <u>Site Map</u>	Firmware Version: 1.00.378		
D-Lini	K								
DVG-2101SP	SETUP	STATUS	HELP						
Device Info	NETWORK STATU	IETWORK STATUS							
VoIP Status	You can use the Network current network conditio	ort to realize the	This is a list of the D-Link VoIP device's routing table.						
VPN Status	ROUTING TABLE								
NetWork Statistic	Display the routing information this device uses to choose the interface/gateway when delivering the datagram to a remote network.								
Logout									
	Flags	Network Destination	Netmask	Gateway	,	Interface			
	U	192.168.1.0	255.255.255.0	0.0.0.0		WAN			
	U	192.168.40.0	255.255.255.0	0.0.0.0		WAN			
	UG	0.0.0.0	0.0.0.0	192.168.4	0.254	WAN			
	NETWORK STATIS View all active TCP cone First Page	SITCS ctions and the TCP and UDP p .ast Page Previous	orts on whiche the	device is liste	ning.				
	Protocol	Message							
	Ip	6383 total packets received							
	Ip	0 forwarded							
	Ip	0 incoming packets discarded							
	Ip	5896 incoming packets delive	ered						
	Ip	6576 requests sent out							
	Icmp	0 ICMP messages received							
	Icmp	0 input ICMP message failed	•						
	Icmp	ICMP input histogram:							
	Icmp	0 ICMP messages sent							
	Icmp	0 ICMP messages failed							
BROADBAND									

System Log You can view all the records of events that have occurred on the VoIP Adapter.

Product Page: DVG-2101	SP				▶ <u>Site Map</u>	Firmware Version: 1.00.378		
D-Lin	k							
DVG-2101SP	SETUP	A	DVANCED	MAINTENANCE	STATUS	HELP		
Device Info	VIEW LOG					Helpful Hints		
VoIP Status	View any attempts th	at have been r	nade to gain access	to your network		Display the captured log messages of the D-Link		
CDR						VoIP device activities, these captured log		
VPN Status						messages might be useful for trouble shooting and		
NetWork Statistic	First Page	Last Page	Previous	Next		monitoring.		
System Log	Page 1 of 4	Page 1 of 4						
Logout	Date/Time	Severity		Message		More		
	Nov 30 10:00:22	info	Lease of 192.168	8.40.119 obtained, lease time 8	6400			
	Nov 30 10:00:22	debug	Sending select fo	r 192.168.40.119				
	Nov 30 10:00:22	debug	Sending discover					
	Nov 30 10:00:20	info	udhcp client (v0.	9.8) started				
	May 16 11:58:19	info	root from 192.16	8.40.99 login success				
	May 16 11:58:19	info	root from 192.16	8.40.99 login success				
	May 16 11:58:10	info	root from 192.16	8. 1. 122 logout				
	May 16 11:58:09	info	root from 192.16	8.1.122 login success				
	May 16 11:54:21	info	root from 192.16	8.1.122 login success				
	May 16 11:53:30	info	root from 192.16	8.1.122 login fail				
BROODBODD								

Help

Help Menu

This help menu provides some information to help you know more details about the VoIP Adapter. Select the item which you want to view to display the description.

Product Page: DVG-2101SP				Site Map	Firmware Version: 1.00.378
D-Link	~				
DVG-2101SP	SETUP	ADVANCED	MAINTENANCE	STATUS	HELP
Menu	HELP MENU				
Setup	<u>Setup</u> Advanced				
Maintenance	<u>Maintenance</u> <u>Status</u>				
Status					
	SETUP HELP				
	<u>Wizard</u> Internet Setup				
	O <u>Connect</u> O <u>MAC Ad</u>	tion Type dress Clone			
	O Static Re VoIP Setup O SIP According	outing			
	Time and Date	look			
	ADVANCED HELP				
	VoIP Settings O Call Fea	hire			
	o <u>SIP Sett</u> o <u>Digit Pla</u>	<u>ing</u> n			
	O Digit Ma O Dial Pret	<u>P</u> fix			
	DNS	etting			
	Dynamic DNS Qos VPN				
	• <u>vriv</u>				
	MAINTENANCE HE	LP			
	System				
	O Configu O Auto Pro	<u>ovision</u>			
	Eirmware Update	ting E			
	<u>Access Controls</u> <u>Ping Test</u>				
	STATUS HELP				
	Device Info VoIP Status				
	<u>CDR</u> <u>VPN Status</u> Network Status				
	<u>System Log</u>				
BROODBODD					
BROADBAILD					

Configuration through IVR

The VoIP Adapter provides a convenient IVR (Interactive Voice Response) function to help you query and configure the basic settings of the VoIP Adapter with a phone and without turning on a PC. To access IVR mode, plug a phone into the VoIP Adapter and then pick up the handset and press * (star) 666 # (pound). You will hear voice instructions through the handset. Follow the system's instructions to set up or query the VoIP Adapter's settings. To exit IVR mode, simply hang up the phone. The IVR function only supports DHCP and Static mode configuration, other connection types need to be done via the web configuration interface. If you pick up the handset and can not hear a tone or press *666# without IVR response, please press the Reset button for 1-2 seconds and try again. Please refer to the Rear Panel diagram on page 5 of this manual for the Reset button location.



Frequently Asked Questions

Q: Why can't I access the web-based configuration utility?

A: When entering the IP address of the VoIP Adapter (192.168.1.150), you are not connecting to a website on the Internet and don't have to be connected to the Internet. The device has the utility built into a ROM chip in the device itself. Your computer must be on the same IP subnet to connect to the web-based utility.

Make sure you have an updated Java-enabled web browser. We recommend the following: Internet Explorer 6.0 or higher Netscape 8 or higher

Mozilla 1.7.12 (5.0) or higher Opera 8.5 or higher Safari 1.2 or higher (with Java 1.3.1 or higher) Camino 0.8.4 or higher Firefox 1.5 or higher

Verify physical connectivity by checking for solid link lights on the device. If you do not get a solid link light, try using a different cable or connect to a different port on the device if possible. If the computer is turned off, the link light may not be on.

Disable any Internet security software running on the computer. Software firewalls such as Zone Alarm, Black Ice, Sygate, Norton Personal Firewall, and Windows® XP firewall may block access to the configuration pages.

Configure your Internet settings:

Go to Start > Settings > Control Panel. Double-click the Internet Options Icon. From the Security tab, click the button to restore the settings to their defaults.

Click the Connection tab and set the dial-up option to Never Dial a Connection. Click the LAN Settings button. Make sure nothing is checked. Click OK.

Go to the Advanced tab and click the button to restore these settings to their defaults. Click OK three times.

Close your web browser (if open) and open it.

Access the web-based utility. Open your web browser and enter the IP address of your DVG-2101SP in the address bar. This should open the login page for the web-based utility.

If you still cannot access the configuration, unplug the power to the VoIP Adapter for 10 seconds and plug it back in. Wait about 1 minute and try accessing the configuration. If you have multiple computers, try connecting using a different computer.

Q: What can I do if I forgot my password?

A: If you forgot your password, you must reset your VoIP Adapter. Unfortunately this process will change all of your settings back to the factory defaults.

To reset the VoIP Adapter, locate the reset button on the rear panel of the unit. With the VoIP Adapter powered on, use a paperclip to hold the button down for 10 seconds. Release the button and the VoIP Adapter will go through its reboot process. Wait about 1 minute to access the VoIP Adapter. The default IP address is 192.168.1.150. When logging in, the username is admin and leave the password box empty.

Q: How can I check if my IP address is correct for logging in?

A: After you install your DVG-2101SP, by default, the TCP/IP settings should be set to obtain an IP address from a DHCP server automatically. To verify your IP address, please follow the steps below. Click on Start > Run. In the run box type cmd and click OK.

At the prompt, type ipconfig and press Enter.

This will display the IP address, subnet mask, and the default gateway of your adapter.

If the address is 0.0.0.0, check your adapter installation, security settings, and the settings on your Adapter. Some firewall software programs may block a DHCP request on newly installed adapters.

If you are connecting to a wireless network at a hotspot (e.g. hotel, coffee shop, airport), please contact an employee or administrator to verify their wireless network settings.

Q: How can I set my computer to use a static IP address?

A: If you are not using a DHCP capable gateway/router, or you need to assign a static IP address, please follow the steps below:

1. Windows® XP: Click on Start > Control Panel > Network Connections.

Windows® 2000: From the desktop, right-click My Network Places > Properties.

- 2. Right-click on the Local Area Connection which represents your network adapter and select Properties.
- 3. Highlight Internet Protocol (TCP/IP) and click Properties.

4. Click Use the following IP address and enter an IP address that is on the same subnet as your network or the LAN IP address on your Adapter.

Example: If the Adapter's LAN IP address is 192.168.1.150, make your IP address 192.168.1.x where x is a number between 1 and 254 but except (192.168.1.1 for example). Make sure that the number you choose is not in use on the network. Set Default Gateway the same as the LAN IP address of your DVG-2101SP (192.168.1.150).

5. Click OK twice to save your settings.

Q: I get a dial tone, but when I dial any phone number I receive a busy signal.

A: The unit may not have been properly configured with the correct account information in the web-based utility.

- 1. Access the web-based utility by entering the IP address: 192.168.1.150 in the Address bar.
- 2. Select the Setup tab and click the SIP Account List submenu.

3. Enter the account information provided by your VoIP service provider, including Display Name, User Name, Authentication Name, Password, Proxy Server (if applicable), Outbound Proxy Server (if applicable), etc.

Q: After setting up the DVG-2101SP, I cannot access the Internet.

A: Make sure that the Ethernet cable from the PC is plugged into the LAN port of broadband modem or router after you finish the configuration of the DVG-2101SP.

Q: Can I use my DVG-2101SP on a MGCP network?

A: No, DVG-2101SP is a SIP device and uses the SIP protocol in handling all of the telephony elements.

Technical Specifications

SIP2 (RFC3261)
SIP1 (RFC2543)
G.711 a/u-law, G.726, G.729A/B, G.723.1
Loop-Start FXS interfaces
Automatic PSTN fallback when power loss
DTMF tone detection/generation
V.21/V.25 Modem/Fax tone detection
Echo Cancellation: G.168
WAN: 10/100 Auto MDI/MDI-X Ethernet ports
IEEE 802.3 10BASE-T Ethernet compliant
IEEE 802.3u 100BASE-TX Fast Ethernet compliant
Voice service is prioritized over the data traffic
(When unit is connected directly to the modem)
Will route all standard protocols
DHCP: Dynamic Host Configuration Protocol server and client
NAT: Network Address Translation
PPPoE: PPP over Ethernet Client
Manage functions through a web-based configuration interface
Power, Provision, Register, WAN, Phone and Line
3.5" x 3" x 1.875" (W x D x H)
One Ethernet port (WAN)
One Phone port (FXS)
One Phone port (PSTN backup)
DC Input: 12VDC/1.5A
0 to 50°C
-10 to 55°C
5% - 95% non-condensing
C-Tick/CUL
FCC Class B
CE Class B

FCC EMI Statement:

This equipment has been tested and found to comply with the limits for a Class B digital device, pursuant to Part 15 of the FCC Rules. These limits are designed to provide reasonable protection against harmful interference in a residential installation. This equipment generates, uses and can radiate radio frequency energy and, if not installed and used in accordance with the instructions, may cause harmful interference to radio communications. However, there is no guarantee that interference will not occur in a particular installation. If this equipment does cause harmful interference to radio or television reception, which can be determined by turning the equipment off and on, the user is encouraged to try to correct the interference by one of the following measures:

- Reorient or relocate the receiving antenna.
- Increase the separation between the equipment and receiver.
- Connect the equipment into an outlet on a circuit different from that to which the receiver is connected.
- Consult the dealer or an experienced radio/TV technician for help.

FCC Caution:

Any changes or modifications not expressly approved by the party responsible for compliance could void the user's authority to operate this equipment. This device complies with Part 15 of the FCC Rules. Operation is subject to the following two conditions: (1) This device may not cause harmful interference, and (2) this device must accept any interference received, including interference that may cause undesired operation.

Technical Support

You can find software updates and user documentation on the D-Link website.

D-Link Australia

Building A, Level 3 11 Talavera Road North Ryde, NSW 2113 Sydney, Australia

Phone: 02 8899 1800 Fax: 02 8899 1868

Australia: 1300 766 868 New Zealand: 0800 900 900

URL: www.dlink.com.au www.dlink.co.nz

E-MAIL: support@dlink.com.au support@dlink.co.nz