Omni.Net Lite

ISDN Terminal Adapter

User's Guide

Version 1.2 5/2005



Copyright

Copyright © 2005 by ZyXEL Communications Corporation.

The contents of this publication may not be reproduced in any part or as a whole, transcribed, stored in a retrieval system, translated into any language, or transmitted in any form or by any means, electronic, mechanical, magnetic, optical, chemical, photocopying, manual, or otherwise, without the prior written permission of ZyXEL Communications Corporation.

Published by ZyXEL Communications Corporation. All rights reserved.

Disclaimer

ZyXEL does not assume any liability arising out of the application or use of any products, or software described herein. Neither does it convey any license under its patent rights nor the patent rights of others. ZyXEL further reserves the right to make changes in any products described herein without notice. This publication is subject to change without notice.

Trademarks

ZyNOS (ZyXEL Network Operating System) is a registered trademark of ZyXEL Communications, Inc. Other trademarks mentioned in this publication are used for identification purposes only and may be properties of their respective owners.

Federal Communications Commission (FCC) Interference Statement

This device complies with Part 15 of FCC rules. Operation is subject to the following two conditions:

- This device may not cause harmful interference.
- This device must accept any interference received, including interference that may cause undesired operations.

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 commercial environment. 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.

If this equipment does cause harmful interference to radio/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 or more of the following measures:

- Reorient or relocate the receiving antenna.
- Increase the separation between the equipment and the 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.

Notice 1

Changes or modifications not expressly approved by the party responsible for compliance could void the user's authority to operate the equipment.

This Class B digital apparatus complies with Canadian ICES-003. Cet appareil numérique de la classe B est conforme à la norme NMB-003 du Canada.

Certifications

- 1 Go to <u>www.zyxel.com</u>.
- **2** Select your product from the drop-down list box on the ZyXEL home page to go to that product's page.
- **3** Select the certification you wish to view from this page.

Safety Warnings

For your safety, be sure to read and follow all warning notices and instructions.

- To reduce the risk of fire, use only No. 26 AWG (American Wire Gauge) or larger telecommunication line cord.
- Do NOT open the device or unit. Opening or removing covers can expose you to dangerous high voltage points or other risks. ONLY qualified service personnel can service the device. Please contact your vendor for further information.
- Use ONLY the dedicated power supply for your device. Connect the power cord or power adaptor to the right supply voltage (110V AC in North America or 230V AC in Europe).
- Do NOT use the device if the power supply is damaged as it might cause electrocution.
- If the power supply is damaged, remove it from the power outlet.
- Do NOT attempt to repair the power supply. Contact your local vendor to order a new power supply.
- Place connecting cables carefully so that no one will step on them or stumble over them. Do NOT allow anything to rest on the power cord and do NOT locate the product where anyone can walk on the power cord.
- If you wall mount your device, make sure that no electrical, gas or water pipes will be damaged.
- Do NOT install nor use your device during a thunderstorm. There may be a remote risk of electric shock from lightning.
- Do NOT expose your device to dampness, dust or corrosive liquids.
- Do NOT use this product near water, for example, in a wet basement or near a swimming pool.
- Make sure to connect the cables to the correct ports.
- Do NOT obstruct the device ventilation slots, as insufficient airflow may harm your device.
- Do NOT store things on the device.
- Connect ONLY suitable accessories to the device.

ZyXEL Limited Warranty

ZyXEL warrants to the original end user (purchaser) that this product is free from any defects in materials or workmanship for a period of up to two (2) years from the date of purchase. During the warranty period, and upon proof of purchase, should the product have indications of failure due to faulty workmanship and/or materials, ZyXEL will, at its discretion, repair or replace the defective products or components without charge for either parts or labor, and to whatever extent it shall deem necessary to restore the product or components to proper operating condition. Any replacement will consist of a new or re-manufactured functionally equivalent product of equal value, and will be solely at the discretion of ZyXEL. This warranty shall not apply if the product is modified, misused, tampered with, damaged by an act of God, or subjected to abnormal working conditions.

Note

Repair or replacement, as provided under this warranty, is the exclusive remedy of the purchaser. This warranty is in lieu of all other warranties, express or implied, including any implied warranty of merchantability or fitness for a particular use or purpose. ZyXEL shall in no event be held liable for indirect or consequential damages of any kind of character to the purchaser.

To obtain the services of this warranty, contact ZyXEL's Service Center for your Return Material Authorization number (RMA). Products must be returned Postage Prepaid. It is recommended that the unit be insured when shipped. Any returned products without proof of purchase or those with an out-dated warranty will be repaired or replaced (at the discretion of ZyXEL) and the customer will be billed for parts and labor. All repaired or replaced products will be shipped by ZyXEL to the corresponding return address, Postage Paid. This warranty gives you specific legal rights, and you may also have other rights that vary from country to country.

Online Registration

Register online at www.zyxel.com for free future product updates and information.

Customer Support

Please have the following information ready when you contact customer support.

- Product model and serial number.
- Warranty Information.
- Date that you received your device.
- Brief description of the problem and the steps you took to solve it.

METHOD	SUPPORT E-MAIL	TELEPHONEA	WEB SITE	
LOCATION	SALES E-MAIL	FAX	FTP SITE	
	support@zyxel.com.tw	+886-3-578-3942	www.zyxel.com www.europe.zyxel.com	ZyXEL Communications Corp. 6 Innovation Road II Science Park
(WORLDWIDE)	sales@zyxel.com.tw	+886-3-578-2439	ftp.zyxel.com ftp.europe.zyxel.com	Hsinchu 300 Taiwan
	info@cz.zyxel.com	+420 241 091 350	www.zyxel.cz	ZyXEL Communications
CZECH REPUBLIC	info@cz.zyxel.com	+420 241 091 359		Modranská 621 143 01 Praha 4 - Modrany Ceská Republika
	support@zyxel.dk	+45 39 55 07 00	www.zyxel.dk	ZyXELCommunicationsA/S
DENMARK	sales@zyxel.dk	+45 39 55 07 07		2860 Soeborg Denmark
	support@zyxel.fi	+358-9-4780-8411	www.zyxel.fi	ZyXEL Communications Oy
FINLAND	sales@zyxel.fi	+358-9-4780 8448		00700 Helsinki Finland
	info@zyxel.fr	+33 (0)4 72 52 97 97	www.zyxel.fr	ZyXEL France
FRANCE		+33 (0)4 72 52 19 20		Bat. 1 / C 69760 Limonest France
	support@zyxel.de	+49-2405-6909-0	www.zyxel.de	ZyXEL Deutschland GmbH.
GERMANY	sales@zyxel.de	+49-2405-6909-99		Wuerselen Germany
	support@zyxel.com	+1-800-255-4101 +1-714-632-0882	www.us.zyxel.com	ZyXEL Communications Inc. 1130 N. Miller St. Anabeim
	sales@zyxel.com	+1-714-632-0858	ftp.us.zyxel.com	CA 92806-2001 U.S.A.
	support@zyxel.no	+47 22 80 61 80	www.zyxel.no	ZyXELCommunicationsA/S
NORWAY	sales@zyxel.no	+47 22 80 61 81		0667 Oslo Norway
	support@zyxel.es	+34 902 195 420	www.zyxel.es	ZyXEL Communications
SPAIN	sales@zyxel.es	+34 913 005 345		1°, 28043 Madrid Spain
	support@zyxel.se	+46 31 744 7700	www.zyxel.se	ZyXEL Communications A/S
SWEDEN	sales@zyxel.se	+46 31 744 7701		Sweden

METHOD	SUPPORT E-MAIL	TELEPHONE ^A	WEB SITE		
LOCATION	SALES E-MAIL	FAX	FTP SITE		
	support@zyxel.co.uk	+44 (0) 8702 909090	www.zyxel.co.uk	ZyXEL Communications UK	
UNITED KINGDOM	sales@zyxel.co.uk	+44 (0) 8702 909091 0906 7370001(UK only)	ftp.zyxel.co.uk	Eastern Road, Bracknell, Berkshire, RG12 2XB, United Kingdom (UK)	

a. "+" is the (prefix) number you enter to make an international telephone call.

Table of Contents

Copyright	2
Federal Communications Commission (FCC) Interference Statement	3
Safety Warnings	4
ZyXEL Limited Warranty	5
Customer Support	6
Table of Contents	8
List of Figures	12
List of Tables	14
Preface	16
Chapter 1 Introduction	18
1.1 About your TA	18
1.1.1 TA Features	18
1.1.1.1 ISDN Basic Rate Interface (BRI) Support	18
1.1.1.2 Extensive Analog Phone Support	18
1.1.1.3 Outgoing Data Call Bumping Support	18
1.1.1.4 PPP Multilink	18
1.1.1.5 Bandwidth-On-Demand	19
1.1.1.6 PAP and CHAP Security	19
1.1.1.7 Upgrade Firmware	19
1.1.1.8 Supplementary Voice Features	19
1.1.1.9 Switch Type	19
1.1.2 TA and ISDN Interfaces	19
1.2 TA Hardware and Driver Installation	20
1.3 Top Panel LEDs	20
1.4 Configuration Methods	21
Chapter 2 Using the Utility	24
2.1 Accessing the TA with the Utility	24
2.2 Navigating the Utility	26
2.3 System	27

2.4 RS-232 Port: Local DN (Directory Number) and Call Type	
2.4.1 Introduction to PPP	
2.4.2 Multilink PPP (MP or MLP)	
2.4.3 BACP and BAP	
2.4.4 ISDN Protocols	
2.4.5 Configuring Local DN and Call Type	
2.5 RS-232 Port: PPP/MP Parameters	
2.5.1 Link Authentication	
2.5.1.1 Windows Authentication	
2.5.2 Call Bumping (Bandwidth Release) for Voice Calls	
2.5.3 Bandwidth On Demand (BOD)	
2.5.4 Configuring PPP/MP Parameters	35
2.6 RS-232 Port: V.24 Signal	
2.6.1 RS-232 Flow Control	
2.6.2 Configuring V.24 Signal	
2.7 Incoming Call Screening	
2.8 Outgoing Call Screening	
2.9 Pre-stored Numbers	40
2.10 RS-232 Port: Advance Setting	41
2.11 Analog Port: Local DN and Options	43
2.11.1 MSN (Multiple Subscriber Number) and Subaddress	43
2.11.1.1 Phone Number Assignment Example	43
2.11.2 Caller ID	44
2.11.2.1 Type 1: On-hook Caller ID Transmission	44
2.11.2.2 Type 2: Off-hook Caller ID Transmission	44
2.11.3 Dialing Method	45
2.11.4 Configuring Local DN and Options	45
2.12 Voice Calls	46
2.12.1 Placing a Voice Call	46
2.12.2 Answering a Voice Call	47
2.12.3 Intercom	47
2.12.4 Supplementary Services	47
2.12.4.1 The Flash Key	48
2.12.4.2 Call Hold	48
2.12.4.3 Call Waiting	49
2.12.4.4 Call Transfer	50
2.12.4.5 Three-Way Conference	51
Chapter 3 Using AT Commands	
3.1 AT Commands Overview	50
3.2 Accessing the TA Example	52 50
3.3 AT Commande	52 55

3.3.1 Data Command Sets	55
3.3.2 Command Sets for Analog Ports	59
3.3.3 Debug Command Sets	61
3.4 Configuring Examples with AT Commands	61
3.4.1 Phone Number Assignment	62
3.4.2 Storing Phone Number	62
3.4.3 Setting Ports Priority	63
3.4.4 Disabling Call Waiting	63
3.4.5 Link Authentication	63
3.4.6 Call Bumping (Bandwidth Release) for Voice Calls	64
3.4.7 Bandwidth On Demand (BOD)	64
3.4.8 Incoming Call Screening	64
3.4.8.1 Data Call	64
3.4.8.2 Voice Call	65
3.4.9 Outgoing Call Screening	65
3.4.9.1 Data Call	65
3.4.9.2 Voice Call	65
3.4.10 Caller ID	65
3.5 Result Codes	66

Chapter 4

Using Tone Commands	68
4.1 How to Use Tone Commands with a Telephone Keypad	68
4.1.1 Tone and Ring Patterns	68
4.2 Tone Commands	69
4.3 Configuring Example with Tone Commands	70
4.3.1 Phone Number Assignment	70
4.3.2 Storing Phone Number	71
4.3.3 Setting Ports Priority	71
4.3.4 Disabling Call Waiting	71
4.3.5 Incoming Call Screening	71
4.3.5.1 Voice Call	71
4.3.6 Outgoing Call Screening	71
4.3.6.1 Voice Call	71
4.3.7 Caller ID	71
Chapter 5	
Firmware Upgrade	72
5.1 Upgrading Using the Utility	72
5.2 Upgrading Using HyperTerminal Example	73

Chapter 6	
Troubleshooting	
Index	80

List of Figures

Figure 1 TA and ISDN Interfaces	20
Figure 2 Top Panel	20
Figure 3 ZyXEL Utility Main Screen	26
Figure 4 System	28
Figure 5 Point-to-Point Link Pathway	29
Figure 6 RS-232 Port: Local DN and Call Type 3	31
Figure 7 Windows XP: Dial-up Connection: Property	33
Figure 8 Windows XP: Dial-up Connection Property: Security	33
Figure 9 Windows 98: Dial-up Connection Property: Server Types	34
Figure 10 RS-232 Port: PPP/MP Parameters	35
Figure 11 CTS/RTS Handshake 3	36
Figure 12 RS-232 Port: V.24 Signal	37
Figure 13 RS-232 Port: Incoming Call Screen	39
Figure 14 Analog Port: Outgoing Call Screen	10
Figure 15 Analog Port: Pre-stored Numbers 4	1
Figure 16 RS-232 Port: Advance Setting 4	12
Figure 17 Analog Port: Local DN and Options 4	15
Figure 18 Analog Port: Supplementary Service 5	50
Figure 19 AT Command Example 1 6	32
Figure 20 AT Command Example 2 6	33
Figure 21 AT Command Example 3 6	35
Figure 22 Utility: Navigation Panel	'2
Figure 23 Utility Firmware Upgrade: Open a File	'2
Figure 24 Utility Firmware Upgrade: Confirm	'3
Figure 25 Utility Firmware Upgrade: In Progress	'3
Figure 26 Utility Firmware Upgrade: Restart	'3

List of Tables

Table 1 Top Panel LED Description	. 20
Table 2 Navigation Panel	. 25
Table 3 Screens Summary	. 26
Table 4 ISDN B-channel Protocols	. 30
Table 5 RS-232 Port: Local DN and Call Type	. 31
Table 6 RS-232 Port: PPP/MP Parameters	. 35
Table 7 RS-232 Port: V.24 Signal	. 38
Table 8 RS-232 Port: Advance Setting	. 42
Table 9 Caller ID signaling type	. 44
Table 10 Analog Port: Local DN and Options	. 46
Table 11 Incoming Voice Call Scenarios	. 47
Table 12 Flash Key commands	. 48
Table 13 Examples of AT Command	. 52
Table 14 AT Data Command Sets Requiring an "AT" Prefix	55
Table 15 AT Command Sets for the Analog Port	. 59
Table 16 Debug Command Sets	. 61
Table 17 PPP AT commands	63
Table 18 Call Bumping AT commands	. 64
Table 19 BOD AT commands	. 64
Table 20 Result Codes	. <mark>66</mark>
Table 21 Tone and Ring Patterns	. <mark>68</mark>
Table 22 Tone Commands	. <mark>69</mark>
Table 23 Troubleshooting	. 78



Congratulations on your purchase of the Omni.Net Lite ISDN Terminal Adapter.

Note: Register your product online to receive e-mail notices of firmware upgrades and information at www.zyxel.com for global products, or at www.us.zyxel.com for North American products.

Your TA is easy to install and configure.

About This User's Guide

This manual is designed to guide you through the configuration of your TA for its various applications.

Related Documentation

• Supporting Disk

Refer to the included CD for support documents.

• Quick Start Guide

The Quick Start Guide is designed to help you get up and running right away. They contain hardware installation/connection information.

• ZyXEL Glossary and Web Site

Please refer to www.zyxel.com for an online glossary of networking terms and additional support documentation.

User Guide Feedback

Help us help you. E-mail all User Guide-related comments, questions or suggestions for improvement to techwriters@zyxel.com.tw or send regular mail to The Technical Writing Team, ZyXEL Communications Corp., 6 Innovation Road II, Science-Based Industrial Park, Hsinchu, 300, Taiwan. Thank you.

Syntax Conventions

- "Enter" means for you to type one or more characters. "Select" or "Choose" means for you to use one predefined choices.
- The SMT menu titles and labels are in **Bold Times New Roman** font. Predefined field choices are in **Bold Arial** font. Command and arrow keys are enclosed in square brackets. [ENTER] means the Enter, or carriage return key; [ESC] means the Escape key and [SPACE BAR] means the Space Bar.
- Mouse action sequences are denoted using a comma. For example, "click **Start**, **Settings** and then **Control Panel**" means first click **Start**, then point your mouse pointer to **Settings** and then click **Control Panel**.
- Use "e.g.," as a shorthand for "for instance", and "i.e.," for "that is" or "in other words".

• The Omni.Net Lite ISDN Terminal Adapter may be referred to as the TA in this user's guide.

CHAPTER 1 Introduction

This chapter introduces the key features of the TA.

1.1 About your TA

The Omni.Net Lite is an ISDN Terminal Adapter (TA) with S/T interface, equipped with two phone ports for connecting to analog phones and/or fax devices. It also supports supplementary services such as call hold, call transfer, call waiting and 3-way conference calling. It can transmit data at the rate of up to 230 Kbps through the RS-232 cable to your computer or terminal.

1.1.1 TA Features

The following sections describe features of the TA.

1.1.1.1 ISDN Basic Rate Interface (BRI) Support

The TA supports a single BRI. A BRI offers two 64 Kbps channels, which can be used independently for two destinations or be bundled to speed up data transfer.

1.1.1.2 Extensive Analog Phone Support

The TA is equipped with two standard phone jacks for you to connect analog devices such as telephones and FAX machines. It also supports supplementary services such as call waiting and 3-way calling.

1.1.1.3 Outgoing Data Call Bumping Support

Call bumping is a feature that allows the TA to manage an MP bundle dynamically, dropping or reconnecting a B-channel in a bundle when necessary.

1.1.1.4 PPP Multilink

The TA can bundle multiple links in a single connection using PPP Multilink Protocol (MP/ MLP/MLPPP). The number of links can be either statically configured or dynamically managed based on traffic demand.

1.1.1.5 Bandwidth-On-Demand

The TA dynamically allocates bandwidth by dialing and dropping connections according to traffic demand.

1.1.1.6 PAP and CHAP Security

The TA supports PAP (Password Authentication Protocol) and CHAP (Challenge Handshake Authentication Protocol). CHAP is more secure than PAP; however, PAP is readily available on more platforms.

1.1.1.7 Upgrade Firmware

Upgrade firmware using Terminal Emulation software such as Hyperterminal.

1.1.1.8 Supplementary Voice Features

To take full advantage of the supplementary voice services available through the TA's phone ports, you will need to subscribe to your phone company for them. The TA supports the following supplementary voice features on both of its analog, or POTS (Plain Old Telephone Service), phone ports:

- Call Waiting
- Three Way Calling
- Call Transfer
- Call Forwarding

1.1.1.9 Switch Type

The TA supports the DSS-1 switch type.

1.1.2 TA and ISDN Interfaces

The TA comes with an S/T interface that connects to an NT-1 (Network Termination 1) device (supplied by the phone company in Europe).

In the following figure:

- All devices that are designed for ISDN are designated Terminal Equipment 1 (TE1).
- All other communication devices that are not ISDN capable, but have a POTS telephone interface (also called the R interface), including ordinary analog telephones, FAX machines, and modems, are designated Terminal Equipment 2 (TE2).
- The Terminal Adapter (TA) connects a TE2 device to an ISDN S/T bus.

Figure 1 TA and ISDN Interfaces



Note: Do not connect this TA to a U interface.

1.2 TA Hardware and Driver Installation

Follow the instructions in the Quick Start Guide to install the driver and make hardware connections.

1.3 Top Panel LEDs

Figure 2 Top Panel



The following table describes the LEDs on the top panel.

 Table 1
 Top Panel LED Description

LED	STATUS	DESCRIPTION
POWER	On	The TA is receiving power.
	Off	The TA is not receiving power.

LED	STATUS	DESCRIPTION
LNK	On	The TA has a successful connection to a local ISDN switch.
	Blinking	The TA is attempting to make a local connection.
	Off	The TA is not connected to the local ISDN switch or the local connection failed.
B1	On	The ISDN B1 channel is in use.
	Off	The ISDN B1 channel is idle or not established.
B2	On	The ISDN B2 channel is in use.
	Off	The ISDN B2 channel is idle or not established.
AA	On	The TA is in automatic answer mode.
	Blinking	The TA is ringing.
	Off	The TA is not in automatic answer mode.
СР	On	Data compression is not supported at the time of writing. Data compression is being used over the B-channel(s). Compression types are Hi/fn LZS (formerly Stac) for PPP connections, and V.42bis for V.120 or X.75 connections.
	Off	Data compression is not activated.
DTR (Data Terminal Ready)	On	The DTE (Data Terminal Equipment) is your computer or terminal that is connected to the TA TO DTE port. DTR indicates a DTE is ready for communication by RS-232 signal.
	Off	The DTE is not ready for communication with the TA via a RS-232 link.
DSR (Data Set	On	The TA is ready for communication with the DTE.
Ready)	Off	The TA is not ready for communication with the DTE.
RTS (Request To Send)	On	The DTE is ready to receive data. The RTS signal is used in hardware handshaking.
	Off	The DTE is not ready to receive data.
CTS (Clear To	On	The TA is ready to receive data.
Send)	Off	The TA is not ready to receive data.
тх	On	The DTE is sending data to the TA.
	Off	The DTE is not sending data to the TA.
RX	On	The DTE is receiving data from the TA.
	Off	The DTE is not receiving data from the TA.
PHONE 1	On	The telephone connected to this port is in use.
PHONE 2	Off	The telephone connected to this port is idle.

Table 1 Top Panel LED Description

1.4 Configuration Methods

To configure your TA, choose one of the following methods:

• Utility

- AT commands with a terminal emulation program such as Hyperterminal in Windows
- Tone commands using the telephone keypad (for analog ports)

CHAPTER 2 Using the Utility

This chapter shows you how to access and configure the TA using the utility.

2.1 Accessing the TA with the Utility

- 1 Make sure your TA hardware is properly connected (refer to the Quick Start Guide).
- **2** Insert the included CD into your CD-ROM driver.
- **3** Double-click the **ConfigOmniNetLite.exe** icon, as shown below, in the **Utility** folder of your CD.



4 You will see the screen as shown below. Select **Local TA** and click **Next** to access the TA through the data port. Otherwise, select **Remote TA**, enter the phone number and click **Next** to access the remote TA via an ISDN network.

ZyXEL Omni.Net Lite ISDN TA Utility Ver 1.2	
TA selection	
Local TA	
C Remote TA	
Remote TEL:	
<prev next=""> Cancel</prev>	

5 Select **Query from TA** and click **Next** to display the current settings on the TA. Select **Read from file** and click **Next** to open a previously saved configuration file (.CFG) on

C Read from file
ে Query from TA
C Default Settings
<prev next=""> Cancel</prev>

your computer. Otherwise, select **Default Settings** and click **Next** to view the default settings of the TA.

6 After you select **Query from TA** and click **Next**, the following screen appears while the query is processing.

Querying current setting		
	40%	

7 The utility main screen displays, you can see five icons on the navigation panel as shown.

ZyXEL Omn	i.Net Lite	ISDN TA U	tility Ver 1.2	Firmv	ware Ver 1.0
C) Load Default	Page 1	Save	👌 Submit to TA	Upgrade	夫 Exit
E 🌺 TA B Sy	stem		- Local I	Directory Nu	mher

Table 2Navigation Panel

LABEL	DESCRIPTION
Load Default	Click this icon to clears all user-entered configuration information and returns the TA to its factory defaults.
Load	Click this icon to upload a new or previously saved configuration (.CFG) file from your computer to your TA.
Save	Click this icon to to back up (save) the TA's current configuration to a file on your computer.
Submit to TA	Click this icon to save your changes back to the TA.

Table 2Navigation Panel

LABEL	DESCRIPTION
Upgrade	Click this icon to upload firmware to your TA.
Exit	Click this icon at any time to leave the utility without saving.

2.2 Navigating the Utility

This section summarizes how to navigate the utility.

Figure 3 ZyXEL Utility Main Screen

ZyXEL Omni.Net Lite ISDN TA UI	Jtility Ver 1.2 Firmware Ver 1.0	
<mark>₿</mark> Coad Default Load Save	Submit to TA Upgrade Exit	
Load Default Load Save Image: System RS232 Port Imag	Local Directory Number Address CLIP and CLIR selection CLIP (pass self DN to destination site) Call Type HDLC transparent X.75 PPP or MLP V.120 X.75 Setup Transmitting Frame Size 254 (bytes)	

The following table describes the menus listed on the left.

Table 3	Screens	Summary
---------	---------	---------

MENU	SUB-MENU	FUNCTION
System		Use this screen to choose your country code.
RS232 Port		Use these screens under this menu to configure the data port.
	Local DN and Call Type	Use this screen to assign a phone number to data calls and select call type.

MENU	SUB-MENU	FUNCTION
	PPP/MP Parameters	Use this screen to configure PPP/MP settings.
	V.24 Signal	Use this screen to define how to treat data control signal.
	Incoming Call Screen	Use this screen to set restrictions on incoming data calls.
	Outgoing Call Screen	Use this screen to set restrictions on outgoing data calls
	Pre-stored Numbers	Use this screen to store up to five phone numbers for data calls on the TA.
	Advance Setting	Use this screen to configure the advanced settings, including charge information.
Analog Port		Use these screens under this menu to configure the analog ports.
	Local DN and Options	Use this screen to assign phone number(s) to voice calls, set port priority, caller ID service types and dialing method.
	Incoming Call Screen	Use this screen to set restrictions on incoming voice calls.
	Outgoing Call Screen	Use this screen to set restrictions on outgoing voice calls.
	Pre-stored Numbers	Use this screen to store up to five phone numbers for voice calls on the TA.
	Supplementary Service	Use this screen to enable or disable the call waiting feature.

Table 3Screens Summary

2.3 System

Click **System** to display the screen as shown next. Select your country from the **Country Selection Code** drop-down list. The default is **Other**.

Load Default Load	📲 👌 Save Submit to TA	Vpgrade Exit	
TA System RS232 Port Local DN and PPP/MP Par. V.24 Signal Pre-stored NL Advance Set Advance Set Local DN and Local DN and Local DN and Local DN and Call Pre-stored NL Code DN and Pre-stored NL Code DN and Supplementa	Country Selection	Code	

Figure 4 System

2.4 RS-232 Port: Local DN (Directory Number) and Call Type

This section shows you how to configure the data call number and type.

2.4.1 Introduction to PPP

Point-To-Point Protocol (PPP) is designed for simple one-to-one links. These links provide full-duplex, simultaneous, bi-directional operation, and are assumed to deliver packets in order. PPP is intended to provide a common solution for easy connection for a wide variety of hosts, bridges and routers.

In the process of configuring, maintaining and terminating the point-to-point link, PPP goes through several distinct phases, as specified in the following simplified state diagram:





2.4.2 Multilink PPP (MP or MLP)

There are two B channels in basic rate ISDN. This offers the possibility of opening multiple simultaneous channels between systems giving users additional bandwidth on demand. Multilink PPP is a method for bundling both B-channels into one PPP link for higher throughput.

Multilink PPP must be negotiated in the establish-link phase by both peers. If the negotiation is successful, the second link will be dialed after the first link reaches network phase. In some countries the directory number is not the same for both channels. For dialing the second B-channel with a different directory number, both numbers must be obtained before dialing out unless the peer supports BACP and BAP, which will be described later in this chapter. Multilink PPP is described in RFC-1990.

2.4.3 BACP and BAP

Bandwidth Allocation Control Protocol (BACP) and Bandwidth Allocation Protocol (BAP) is used for call request and link drop under Multilink PPP.

BACP is negotiated during the network phase. Without BACP/BAP, the directory numbers of both B-channels must be specified before dialing out. In some cases, it is not possible for the ISP to support more than one chassis that is capable of bundling both channels at the same time. With BACP/BAP, the second B-channel directory number can be obtained while requesting a call. The second directory number is not necessary any more.

2.4.4 ISDN Protocols

HDLC (High-Level Data Link Control) is one of the most common data link (layer 2) protocols. It supports both half duplex and full duplex communication lines, point to point (peer to peer) and multi-point networks, and switched or non-switched channels. It permits synchronous, code-transparent data transmission.

The TA supports the following protocols.

ISDN PROTOCOLS	DESCRIPTION
V.110	V.110 is most popular in Japan or Europe.
V.120	V.120 is most commonly used in North America.
X.75	X.75 was originally designed for packet-switched signaling systems in public networks to provide data transmission services. However, it is now also used as the link layer for telematic services (as defined in T.90) in ISDN. These services include both ISDN circuit-switched mode (DTE-DTE communication) and ISDN packet-switched mode (DTE-DCE communication). DCE (Data Communication Equipment) is a device, such as a modem, that converts data between different interfaces (digital and analog for example) and exchanges data with the DTE.
PPP and MLP	PPP allows B-channels bundling and bandwidth release and is the most widely used protocol for Internet access.

Table 4	ISDN	B -channel	Protocols
	10DIN	D-channer	1 10100013

2.4.5 Configuring Local DN and Call Type

Click Local DN and Call Type under RS-232 Port to open the screen as shown next.

Image: System System Image: System RS232 Port Image: System Image: System Image: System RS232 Port Image: System Image: System Image: System <	ad Default Load Save	Submit to TA Upgrade Exit
CLIP and CLIR selection CLIP (pass self DN to destination site) CLIP (pass self DN tod	With the system Image: System	Local Directory Number Address SubAddress
Incoming Call Screen Dutgoing Call Screen Pre-stored Numbers Supplementary Service Value Call Type HDLC transparent X.75 Pre-stored Numbers PPP or MLP V.120		CLIP and CLIR selection CLIP (pass self DN to destination site)
		Call Type C HDLC transparent C X.75 PPP or MLP C V.110 C V.120
X.75 Setup Transmitting Frame Size 254 ♀ (bytes)		X.75 Setup Transmitting Frame Size 254 ♀ (bytes)

Figure 6 RS-232 Port: Local DN and Call Type

The following table describes the labels in this screen.

LABEL	DESCRIPTION		
Local Directory Number			
Address	Enter the phone number of the data port.		
SubAddress	Enter the sub-address of the data port.		
CLIP and CLIR Selection	CLIP (Calling Line Identity Presentation) displays the phone number of the caller on the callee's telephone. CLIR (Calling Line Identity Restriction) allows you to hide your phone number when you make a call.		
	Select CLIP (pass self DN to destination site) to allow the ISDN switch to pass your phone number to the destination site.		
	Select CLIR (don't pass self DN to destination site) to disable the CLIP service.		

 Table 5
 RS-232 Port: Local DN and Call Type

LABEL	DESCRIPTION
Call Type	Sets the calling protocol for the your TA. The protocol selected should match the protocol in use by the device or network your ISDN TA is calling.
	Select HDLC transparent to apply HDLC on B-channels and allow peer to peer communication similar to what a modem does.
	Select X.75 or V.120 to do CRC (Cyclic Redundancy Check) error detecting on B-channel. X.75 performs error correction using LAPB (Link Access Procedure Balanced) and V.120 uses LAPD (Link Access Procedure on the D Channel).
	Select PPP or MLP for modem-like dial-up connection to the Internet and/or combining two B-channels into one.
	Select V.110 to allow rate adaption between sync 64 kbps B-channel and async 9.6 - 115.2 kbps DTE.
X.75 Setup	
Transmitting Frame	Select a value between 128 and 2048 to change the sending frame size.
Size (bytes)	The default sending frame size is 254 bytes.

2.5 RS-232 Port: PPP/MP Parameters

This section shows you how to configure PPP or MP parameters.

2.5.1 Link Authentication

After a link is established, it is necessary to authenticate the peer for security reasons. There are two popular authentication methods. One is Password Authentication Protocol (PAP) and the other is Challenge Handshake Authentication Protocol (CHAP). PAP is less secure because it transmits the username/password in plain text form. Unlike PAP, CHAP transmits the username/password in encrypted form. Some ISPs may support CHAP as the only method for authentication.

2.5.1.1 Windows Authentication

For those applications that do not support CHAP, the TA converts CHAP into PAP. It is recommended that you select **Allow unsecured password** in the **Dial-up Connection Property Security** screen on Windows 2000/XP or do not select the **Required encrypted password** checkbox in the **Dial-up Connection Property Server Types** screen on Windows 95/98 by right-clicking on the dial-up connection icon (see the figures below), since Microsoft uses MS-CHAP for username and password encryption but MS-CHAP is not supported by most ISPs.



Figure 7 Windows XP: Dial-up Connection: Property

Figure 8 Windows XP: Dial-up Connection Property: Security



inet-RS	232			?
General	Server Types	Scripting	Multilink	
Туре	of Dial-Up <u>S</u> erver	2		
PPP	Internet, Windov	vs NT Ser	ver, Windows	98
_ Adv	anced options:			
N	Log on to netwo	ork		
V	Enable software	e <u>c</u> ompres	sion	
(Г	Require encryp	ted passw	ord	
	Require <u>d</u> ata er	ncryption		
Г	<u>R</u> ecord a log file	e for this c	onnection	
_ Allo	wed network prot	ocols:		
V	<u>N</u> etBEUI			
1	IPX/SPX Comp	atible		
1	ICP/IP		TCP/IP Sett	ings
		ſ	ОК	Cancel

Figure 9 Windows 98: Dial-up Connection Property: Server Types

If you choose MLP 128K (dynamic or fixed) in the utility (see Figure 10 on page 35) or using the ATC0=8CI=1 command, the TA needs the username and password in plain text form to establish the connection. Do not select **Enable CHAP** in the **PPP/MP Parameters** screen.

CHAP is described in RFC-1994 and PAP is described in RFC-1334.

2.5.2 Call Bumping (Bandwidth Release) for Voice Calls

While dynamic Multilink PPP is active and you place or answer a voice call from a device that is attached to one of the **PHONE** ports, the TA automatically drops one of the channels and assigns it to voice calls. If you subscribe to the call waiting service from your local telephone company, the TA will also drop one of the channels and answer the incoming call. Once a voice call ends, the TA automatically re-establishes the channel depending on the BOD rule.

2.5.3 Bandwidth On Demand (BOD)

In multilink PPP mode, use BOD to bring up or drop a B-channel depending on current data traffic. If there is light traffic on the link, one of the channels will be dropped automatically. On the other hand, if only one B-channel is used and data traffic is high enough, BOD will bring up the second B-channel to increase the bandwidth of the data link.

The BOD function is only effective on your TA.

For BOD, there are three parameters to set high/low threshold and persist time. Traffic utilization is measured in the ISDN link. Highly compressible data may not generate enough traffic to start the second B-channel. Both add-persist time and subtract-persist time must be set to activate BOD.

Note: The high threshold must be greater than the low threshold, otherwise, both of them will be reset to the default.

2.5.4 Configuring PPP/MP Parameters

Click **PPP/MP Parameters** under **RS-232 Port** to display the screen as shown next.

Figure 10 RS-232 Port: PPP/MP Parameters

ZyXEL Omni.Net	Lite ISDN TA Ut	ility Ver 1.2	Firmv	vare Ver 1.0	
Load Default Lo	ad Save	Submit to TA	Upgrade	Exit	
	ort IDN and Call Type MP Parameters Signal ning Call Screen tored Numbers nce Setting ort IDN and Options ning Call Screen oing Call Screen tored Numbers lementary Service	 Enable Enable ✓ Enable Maximum 64 kbp Threshol ADD 3 Persist ti 24 	Fixed MP (CHAP Call Bumpi throughput s (1 B-chan d of data lo 5 \$ Kbp me for moni \$ 5 \$	fixed 128K) ng (dynamic E : (1 or 2 B-cha nel)	bandwidth allocation for voice call) annel) /DROP one channel DROP 25 \$ Kbps (0 - 64) old to add or drop one channel

The following table describes the labels in this screen.

Table 6	RS-232 Port:	PPP/MP	Parameters	

LABEL	DESCRIPTION	
Enable Fixed MP (fixed 128K)	Select this option to combine both B-channels into one PPP link for data calls and always has the higher transmission rate.	
Enable CHAP	Select this option to have your TA use CHAP for link authentication with the username and/or password.	
Enable Call Bumping	Select this option to allow bandwidth release for voice calls.	
Maximum throughput (1 or 2 B-channel)	This field is not available if you select Enable Fixed MP.	
	Select 64 kbps (1 B-channel) to activate only one B-channel and transmit data at 64 kbps.	
	Select 128 kbps (2 B-channel) if if you want to bundle two B-channels to increase the data transfer rate. This is called multilink.	
LABEL	DESCRIPTION	
---	--	
Threshold of data loading to ADD/DROP one channel	This field is not available if you select Enable Fixed MP . Set the maximum threshold in the ADD field and the minimum threshold in the DROP field. The TA adds the second B-channel when traffic is above the threshold and drops the second B-channel when traffic is below the threshold.	
Persist time for monitoring threshold to add or drop one channel	This field is not available if you select Enable Fixed MP . Set the time interval between calculating traffic load. The unit of a value is 5 seconds.	

Table 6 RS-232 Port: PPP/MP Parameters

2.6 RS-232 Port: V.24 Signal

The V.24 signal screen describes ways to treat data control signal for data transmitted via the RS-232 serial interface of the TA. You can also use this screen to configure how to control data flow between the computer and the TA.

2.6.1 RS-232 Flow Control

RS-232 Flow control is used to control data flow between devices, such as two modems or a computer and a modem, using a serial connection. Standard flow control protocols include hardware (RTS/CTS) flow control and software (XON/XOFF) flow control.

The RTS (Request To Send) and CTS (Clear to Send) lines of a RS-232 serial cable signal flow control handshaking. In the following example, **A** is a TA and **B** is a computer. For upstream traffic from **B** to **A**, when device **A**'s receive buffer is nearly full, it lowers the CTS line (the **CTS** LED is off) to signal device **B** to stop sending data. When **A** is ready for new incoming data, it raises the CTS signal (the **CTS** LED is on) to resume transmissiom. For downstream traffic from **A** to **B**, **B** sets the RTS line to have **A** stop or restart sending data.





Flow control can also be done using the XON character (by pressing Ctrl-Q on the keyboard) and XOFF characters (by pressing Ctrl-S) in a data stream. When a device receives an XOFF signal from another device, it suspends or stops data transmitting to that device until receiving an XON (Ctrl-Q) signal from the same device. An XOFF character is sent when the receive buffer is nearly full. An XON signal is sent when the buffer is no longer full. Binary files may contain these signal characters and cannot be sent with XON/XOFF flow control protocol.

Note: The TA and the computer should use the same flow control protocol.

2.6.2 Configuring V.24 Signal

Click V.24 Signal under RS-232 Port to display the screen as shown next.

Figure 12 RS-232 Port: V.24 Signal

Load Default Load Save	Submit to TA Upgrade Exit
 → TA → System → RS232 Port → Local DN and Call Type → PPP/MP Parameters → → V.24 Signal 	Data Carrier Detect Control ON only if in communication
Incoming Call Screen Dutgoing Call Screen Dutgoing Call Screen Pre-stored Numbers D Advance Setting Analog Rott	Data Terminal Ready Action ON to OFF, disconnect the call (108/2)
 → Local DN and Options → Incoming Call Screen → Outgoing Call Screen → Pre-stored Numbers 	Flow Control Setting CTS/RTS flow control
🧁 🖹 Supplementary Service	Data Set Ready Follow DTR signal

The following table describes the labels in this screen.

Table 7RS-232 Port: V.24 Signal

LABEL	DESCRIPTION	
Data Carrier Detect Control	Data Carrier Detect (DCD) is a serial communication signal. It indicates whether or not the TA is connected to a remote modem.	
	Select Follow DTR signal to raise the DCD signal and have the LED turn on when the DTR signal is raised and the DTR LED is on and vice versa.	
	Select On only if in communication to raise the DCD signal and have the LED turn on only when a data call is established.	
Data Terminal Ready Action	DTR indicates whether or not the computer or terminal is ready for data transmission.	
	Select ON to OFF, disconnect the call (108/2) to have the computer or terminal disconnect the current data call if the DTR signal is lowered and the LED is off.	
	Select Ignore DTR status so that the TA will not follow the behavior of the DTR signal to drop a call; you then need to use the ATH command to disconnect it.	
	Select OFF to ON, make a call, On to OFF, clear the call (108/1) to have the TA dial out a call using the pre-stored phone number when the DTR LED turns on and disconnect the call when it is off.	
Flow Control Setting	Select the flow control method from the drop-down list. Options are Disable flow control, CTS/RTS flow control and XON/XOFF flow control.	
Data Set Ready	DSR indicates whether or not the TA is ready for data transmission.	
	Select Follow DTR signal to have the DSR LED turn on when the DTR LED is on and vice versa.	
	Select Follow DCD signal to to have the DSR LED turn on when the DCD LED is on and vice versa.	

2.7 Incoming Call Screening

Use this function to set restrictions on incoming data or voice calls.

In **Incoming Call Screen** under **RS-232 Port** or **Analog Port**, create a list of incoming call phone numbers (with sub-addresses) and configure the **Mode** field to reject or accept the calls with the phone number in the list.

oad Default Load Save	Submit to TA Upgrade Exit	
TA System RS232 Port PPP/MP Parameters Y.24 Signal Incoming Call Screen Outgoing Call Screen Pre-stored Numbers Advance Setting Incoming Call Screen Dutgoing Call Screen Dutgoing Call Screen Outgoing Call Screen Dutgoing Call Screen Outgoing Call Screen Outgoing Call Screen Outgoing Call Screen Pre-stored Numbers Outgoing Call Screen Pre-stored Numbers Supplementary Service	Phone list of incoming call screen Address 0 12345678 1 2 3 4 Mode Reject all incoming calls	SubAddress

Figure 13 RS-232 Port: Incoming Call Screen

2.8 Outgoing Call Screening

Use this function to set restrictions on outgoing data or voice calls, such as allowing or prohibiting a long-distance or international call.

In **Outgoing Call Screen** under **RS-232 Port** or **Analog Port**, create a list of prefix digits for outgoing call and configure the **Mode** field to prohibit or allow the calls with the prefix digits in the list.

oad Default 🛛 Load Save	Submit to TA Upgrade Exit	
Image: System Image: System <td< td=""><td>Phone list of outgoing call screen Address 0 1 2 Mode No any restriction</td><td></td></td<>	Phone list of outgoing call screen Address 0 1 2 Mode No any restriction	

Figure 14 Analog Port: Outgoing Call Screen

2.9 Pre-stored Numbers

To store a phone number for data calls or voice calls in the TA, click **Pre-stored Numbers** under **RS-232 Port** or **Analog Port** and enter the phone numbers (and sub-addresses) in the screen shown as below.



Figure 15 Analog Port: Pre-stored Numbers

2.10 RS-232 Port: Advance Setting

Use this screen to configure data port settings and charge information.

Click Advance Setting under RS-232 Port to display the screen as shown next.

oad Default Load Save	🗞 🧖 🤾 Submit to TA Upgrade Exit
Image: System Image: System <td< td=""><td>Advance options Auto Answer 0 1 (number of rings), 0 : Disable Wait for Conn 60 1 (second) Power on baud rate 115200 (default) 1 Power on attribute 8 data bits, non parity (default) 1 Extended result code basic result code (default) 1 Ring result code no extended information (default) 1 Charge Information Last Call Charge Info. 0 Clear last call charge info Accumulate Charge Info. 0 Clear accumulate charge info</td></td<>	Advance options Auto Answer 0 1 (number of rings), 0 : Disable Wait for Conn 60 1 (second) Power on baud rate 115200 (default) 1 Power on attribute 8 data bits, non parity (default) 1 Extended result code basic result code (default) 1 Ring result code no extended information (default) 1 Charge Information Last Call Charge Info. 0 Clear last call charge info Accumulate Charge Info. 0 Clear accumulate charge info
	Power On Auto-Dialing 0 ÷ 0 : Disable Maximum Retry Count to Redial 0 ÷ 0 : Disable

Figure 16 RS-232 Port: Advance Setting

The following table describes the labels in this screen.

LABEL	DESCRIPTION
Advance Options	
Auto Answer	Set how many ringing messages the TA receives before it answers a data call automatically. Enter 0 to stop the TA answering a data call automatically. When there is an incoming data call, you can press the push button on the top panel (see the Quick Start Guide) or use the ATA command to answer it manually.
Wait for Conn	Enter the number of seconds for the TA to wait for an answer to the data call before dropping it.
Power on baud rate	Configure the initial baud rate after power-on.
Power on attribute	Configure the initial attribute of the data port after power-on.
Extended result code	Select extended result code to show the extended command response or the connect message to the DTE (see Table 20 on page 66). Otherwise, select basic result code (default) .
Ring result code	Select the information you want to display for the incoming calls. Options are no extended information (default), display calling phone number, display user data, display calling phone number and user data and RVS-COM format.
Charge Information	

Table 8	RS-232 Port	: Advance Setting
---------	-------------	-------------------

LABEL	DESCRIPTION
Last call charge info.	This field displays the cost for the last call. This feature is not supported by all ISDN switches.
Clear last call charge info.	Select this checkbox to clear the information in the field above.
Accumulate Charge Info.	This field displays total cost for all calls. This feature is not supported by all ISDN switches.
Clear accumulate charge info.	Select this checkbox to clear the information in the field above.
Power On Auto-Dialing	Select 1 to have the TA automatically dial up a connection after power-on. Select 0 to disable this feature.
Maximum Retry Count to Redial	Enter how many times (from 1 to 10) for the TA to try to dial out a data call if a call can not be established successfully. Enter 0 to disable this feature.

2.11 Analog Port: Local DN and Options

This section shows you how to assign the phone number(s) to voice calls and set port priority.

2.11.1 MSN (Multiple Subscriber Number) and Subaddress

Multiple Subscriber Number (MSN) is where the telephone company gives you more than one number for your ISDN line. You can assign each number to a different port, e.g., the first number to data calls, the second to A/B adapter 1 and so on. Or (DSS1) the telephone company may give you only one number, but allow you to assign your own subaddresses to different ports, e.g., subaddress 1 to data calls and 2 to A/B adapter 1.

2.11.1.1 Phone Number Assignment Example

Different telephone companies deploy different types of switches for ISDN service. Depending on the switch for your particular installation, you will have a different number of telephone numbers.

When your telephone company provides you with more than one phone number (say 5009001 and 5009002), you can connect two analog phones and/or G3 fax devices to the **PHONE 1** and **PHONE 2** ports and assign each number to a port. If you attach one analog phone and one fax machine to the analog ports but don't assign any phone number to either port, both telephone and fax ring when there is an incoming call. That will be difficult to tell whether it is a fax call or voice call.

Follow the steps for the configuration method of your choice to assign the phone numbers. After configuration, the TA will route a call to the port that matches the dialed number.

1 Refer to Section 2.1 on page 24 to open the utility.

- 2 Enter the first number in the Address field beside Port 1 and the second number in the Address field beside Port 2.
- **3** Click **Submit to TA** to save the changes back to the TA.

2.11.2 Caller ID

The caller ID (Caller Identification) information will display on the analog ports. Two caller ID service types are supported by your TA.

2.11.2.1 Type 1: On-hook Caller ID Transmission

If the caller ID protocol uses DTMF (Dual Tone Multi-Frequency) signal, the caller ID information includes the phone number and is sent prior to ringing.

If the caller ID protocol uses FSK (Frequency Shift Keying) signal, the phone number will be sent during ringing.

The caller ID signaling type varies depending on the country.

COUNTRY	COUNTRY CODE	CALLER ID SIGNALING TYPE	RINGING PATTERN
Denmark	252	DTMF	1 Sec On, 2 Sec OFF
France	219	FSK	1 Sec On, 2 Sec OFF
German	237	FSK	1 Sec On, 2 Sec OFF
India	214	DTMF	1 Sec On, 2 Sec OFF
Ireland	235	FSK	0.4 Sec On, 0.2 Sec OFF, 0.4 Sec On, 2 Sec OFF
Italy	236	DTMF	1 Sec On, 2 Sec OFF
Netherland	253	DTMF	1 Sec On, 2 Sec OFF
Norway	245	FSK	1 Sec On, 2 Sec OFF
Portugal	220	DTMF	1 Sec On, 2 Sec OFF
Spain	213	DTMF	1 Sec On, 2 Sec OFF
Sweden	250	DTMF	1 Sec On, 2 Sec OFF
Switzerland	251	DTMF	1 Sec On, 2 Sec OFF
UK	249	FSK	0.4 Sec On, 0.2 Sec OFF, 0.4 Sec On, 2 Sec OFF
Other (default)	216	DTMF	1 Sec On, 2 Sec OFF

 Table 9
 Caller ID signaling type

2.11.2.2 Type 2: Off-hook Caller ID Transmission

The Type 2 service uses only FSK and shows call waiting caller ID. If the service is supported and activated on your TA, the call waiting phone number is displayed when you are on the phone.

2.11.3 Dialing Method

There are two ways to send a message containing the phone number: **EN-BLOC sending** and **OVERLAP sending**.

In En-Bloc mode, the complete phone number is sent in the first signaling message. Users enter all digits of the phone number before a timeout period expires (the call times out if there is inactivity for more than four seconds) or press "#" to indicate the complete phone number has been entered.

OVERLAP sending is similar to what the traditional PSTN (Public-Switched Telephone Network) telephone system uses. **OVERLAP sending** means sending some phone digits in the first signaling message and the rest in subsequent signaling message. Users enter the phone number after a dial tone. The ISDN network is aware of the completion of the phone number automatically. You can not enter the sub-address when using overlap sending.

OVERLAP sending is usually used as the default for voice calls. For data calls, only **EN-BLOC sending** is used.

2.11.4 Configuring Local DN and Options

Click Local DN and Options under Analog Port to display the screen as shown next.

Submit to TA Upgrade Exit	
Local Directory Number Sul Port 1 5009001 Port 2 5009002	bAddress
CLIP and CLIR selection Port 1 CLIP (pass self DN to destination site) Port 2 CLIP (pass self DN to destination site)	•
Device type on analog port Port 1 Telephone, G3 Fax or Modem Port 2 Telephone with Call Waiting caller ID device	
Port priority Both ports are same priority	
Dialing method OVERLAP sending	×
	Submit to TA Upgrade Exit Local Directory Number Address Sul Port 1 5009001 Port 2 Port 2 5009002 Port 1 CLIP and CLIR selection Port 1 CLIP (pass self DN to destination site) Port 2 CLIP (pass self DN to destination site) Port 2 Device type on analog port Port 1 Telephone, G3 Fax or Modem Port 2 Telephone with Call Waiting caller ID device Port priority Both ports are same priority Dialing method OVERLAP sending

Figure 17 Analog Port: Local DN and Options

The following table describes the labels in this screen.

 Table 10
 Analog Port: Local DN and Options

LABEL	DESCRIPTION			
Local Directory Number				
Port 1	Enter the ISDN phone number for voice calls in the Address field and the sub			
Port 2	address in the SubAddress field.			
CLIP and CLIR Selection	CLIP (Calling Line Identity Presentation) displays the phone number of the caller on the callee's telephone. CLIR (Calling Line Identity Restriction) allows the phone number to be hid.			
	Select CLIP (pass self DN to destination site) to allow the ISDN switch to pass your phone number to the destination site.			
	Select CLIR (don't pass self DN to destination site) and the CLIP service will not be available.			
Device type on analog port	To enable the Type 2 service on either analog port, select Telephone with Call Waiting caller ID device .			
	Otherwise, select Telephone, G3 Fax or Modem.			
Port Priority	To set the same priority for both two ports, select Port priority to Both ports are same priority.			
	To set the PHONE 1 port to have the first priority, select Port priority to Analog Port 1 is high priority .			
	To set the PHONE 2 port to have the first priority, select Port priority to Analog Port 2 is high priority .			
Dialing method	Configure how to send the signaling message with the phone number. Options are OVERLAP sending and EN-BLOC sending .			

2.12 Voice Calls

This section covers how to initiate and receive calls over an ISDN through the analog ports on your ISDN TA and introduces supplementary services including call hold, call waiting, call transfer and three-way conference.

Note: You can only use the utility, AT command or Tone command to activate or deactivate call waiting. Use the flash key and the digit keys on the telephone to invoke supplementary services (see Section 2.12.4 on page 47).

2.12.1 Placing a Voice Call

You can use your analog phone connected to the **PHONE 1** and **PHONE 2** ports to dial out a call directly. With the **Quick Dial** feature, you can place an ISDN voice call by dialing "nn#" (where nn=00-04, pre-stored phone number index). You must store up to five phone numbers first by using the utility, AT commands or tone commands. Refer to Section 2.9 on page 40 on how to store a phone number with the utility.

2.12.2 Answering a Voice Call

To answer an incoming call, just pick the receiver when that telephone is ringing.

Voice calls are routed to one or both of the analog ports (**PHONE 1** or/and **PHONE 2**) based on the phone number assignment when they are received. When the same number is assigned to both ports, you can choose the port you want to receive calls by setting the priority of the ports. If two ports have same priority, both ports receive ring signals. See Section 2.11.4 on page 45 on how to configure port priority.

When answering an incoming call, the TA will first be identified if the calling number matches the MSN settings. The Multiple Subscriber Number (MSN) supplementary service enables multiple ISDN numbers to be assigned to a single ISDN BRI line. It allows the caller to select, via the public network, one or more distinct terminals from a variety of terminal choices. Since the TA supports many different communication protocols and two analog adapters, each of these ports can individually be assigned to a different ISDN number (see Section 2.11.1 on page 43 for more information).

The following table shows incoming voice call scenarios.

PHONE 1 MSN	PHONE 2 MSN	PORT PRIORITY	PHONE 1	PHONE 2
Unassigned or Matched	Unassigned or Matched	0	Ring	Ring
Unassigned or Matched	Unassigned or Matched	1	Ring	-
Unassigned or Matched	Unassigned or Matched	2	-	Ring
Matched	Not matched	None	Ring	-
Not matched	Matched	None	-	Ring
Not matched	Not matched	None	-	-

Table 11	Incoming	Voice Call	Scenarios
----------	----------	------------	-----------

2.12.3 Intercom

Your TA can work as an "Intercom" by connecting two phones to **PHONE 1** and **PHONE 2** at the rear panel of your TA.

Please follow the instructions below to operate the Intercom:

- **1** Pick up the telephone handset.
- **2** Dial "1#" and the other local phone will start to ring.

2.12.4 Supplementary Services

Supplementary services such as call hold, call waiting, call transfer, ... are generally available from your telecommunications provider. The TA supports all of the following services:

- Call Hold
- Call Waiting
- Call transfer
- Three-Way Conference
- **Note:** To take full advantage of the supplementary voice services available though the TA's phone ports, you will need to subscribe to the services from your local telephone company.

2.12.4.1 The Flash Key

Flashing means to press the hook for a short period of time (a few hundred milliseconds) before releasing it. On newer telephones, there should be a "flash" key (button) that generates the signal electronically. If the flash key is not available, you can tap (press and immediately release) the hook by hand to achieve the same effect. However, using the flash key is preferred since the timing is much more precise. With manually tapping, if the duration is too long, it may be interpreted as hanging up by the TA.

You can invoke all the supplementary services by using the flash key. Commands for supplementary services are listed in the table below

After pressing the flash key, if you do not issue the sub-command before the default subcommand timeout (2 seconds) expires or issue an invalid sub-command, the current operation will be aborted.

COMMAND	SUB-COMMAND	DESCRIPTION
Flash		Put a current call on hold to place a second call.
Flash	0	Drop the call presently on hold or reject an incoming call which is waiting for answer.
Flash	1	Disconnect the current phone connection and answer the incoming call or resume with caller presently on hold.
Flash	2	 Switch back and forth between the two calls. Put a current call on hold to answer an incoming call. Separate the current three-way conference call into two individual calls (one is on-line, the other is on hold).
Flash	3	Create three-way conference connection.
Flash	4	Transfer the call to another phone.

Table 12Flash Key commands

2.12.4.2 Call Hold

Call hold allows you to put a call (A) on hold by pressing the flash key.

If you have another call, press the flash key and then "2" to switch back and forth between caller A and B by putting either one on hold.

Press the flash key and then "0" to disconnect the call presently on hold and keep the current call on line.

Press the flash key and then "1" to disconnect the current call and resume the hold call.

If you hang up the phone but a caller is still on hold, there will be a remind ring (see Section 4.1.1 on page 68 for more information).

2.12.4.3 Call Waiting

This allows you to place a call on hold while you answer another incoming call on the same telephone (directory) number. By default call waiting is enabled on both telephone ports, but can be disabled on either port.

If there is a second call to a telephone number, you will hear a call waiting tone. Take one of the following actions.

• Reject the second call.

Press the flash key, wait for the sub command tone, press "0".

• Disconnect the first call and answer the second call.

Either press the flash key, wait for the sub command tone and press "1", or just hang up the phone and then answer the phone after it rings.

• Put the first call on hold and answer the second call.

Press the flash key, wait for the sub command tone, press "2".

2.12.4.3.1 Disabling Call Waiting

The call waiting tone will interfere with the data communication. It is highly recommended that you disable the call waiting feature on the port to which a fax or modem is connected.

To disable call waiting on either port, select the **Disable PORT 1/2 call waiting** check box in the **Analog Port Supplementary Services** screen of the utility.

Load Default Load Save	Submit to TA Upgrade Exit
Load Save Image: System Image: System Image: System RS232 Port Image: System Image: System Image: System Image: System Image: System Image: System Image: System Image: System Image: Supplementary Service Image: Supplementary Service	e Subplementary service options Disable PORT 1 call waiting Disable PORT 2 call waiting

Figure 18 Analog Port: Supplementary Service

2.12.4.4 Call Transfer

You can transfer a call to a local phone (connected to the TA) or a remote phone (not connected to the TA) through an ISDN line.

2.12.4.4.1 Call Transfer between Two Analog Ports

To transfer incoming calls (after you answer this call) to a local phone, follow these steps:

- **1** Press the flash key to put the caller on hold.
- **2** When you hear the dial tone, dial "1#" to operate the Intercom.
- **3** After you hear the ring signal or the second party answer it, press the flash key, then "4".
- **4** Hang up the phone.

2.12.4.4.2 Call Transfer to a Remote Phone

To transfer incoming calls (after you answer this call) to a remote phone, follow these steps:

- **1** Press the flash key to put the caller on hold.
- **2** When you hear the dial tone, simply dial the number you want, to transfer the first call.
- **3** After you hear the ring signal or the second party answer it, press the flash key, then "4".

4 Hang up the phone.

2.12.4.5 Three-Way Conference

- **1** When you are on the phone talking to someone, place the flash key to put the caller on hold and get a dial tone.
- **2** Dial a phone number directly to make another call.
- **3** When the second call is answered, press the flash key, wait for the sub-command tone and press "3" to create a three-way conversation.
- **4** Hang up the phone to drop the connection.
- **5** If you want to separate the activated three-way conference into two individual connections (one is on-line, the other is on hold), press the flash key, wait for the sub-command tone and press "2".

CHAPTER 3 Using AT Commands

3.1 AT Commands Overview

An AT Command is a command in asynchronous data format issued by the computer to the ISDN TA through the asynchronous computer-modem interface. AT commands are used to configure and control the TA by typing commands at a computer or terminal keyboard. To send an AT Command from a computer to the TA, you must run a communication software program and the TA must be in command state.

Commands must be written in a specific form in order for the TA to recognize them. A command begins with the letters "AT" or "at", followed by one or more printable characters and then ended by pressing [ENTER], as shown in the following example.

Table 13	Examples o	f AT Command
----------	------------	--------------

COMMAND	OPERATION	RESPONSE
AT <enter></enter>	To verify that the TA is online with your terminal or computer.	ОК
AT 13	T I3 To display the version of the current firmware.	

Note: AT commands can only be issued when the computer or terminal is connected to the TA with an RS-232 cable.

3.2 Accessing the TA Example

To configure the TA by terminal emulation program, you need a terminal emulation program first, such as Hyperterminal in Windows.

- 1 Make sure your TA hardware is properly connected (refer to the Quick Start Guide).
- **2** Open the HyperTerminal program in Windows. For example, in Windows XP click **Start, Programs, Accessories, Communications, HyperTerminal**.

3 Enter a descriptive name and select an icon in the **Connection Description** screen. Click **OK** to display the **Connect To** screen.



4 Select the COM port to which the TA is connected. Click **OK** to continue.

Connect To	? 🛛
SISDN T4	,
Enter details for	the phone number that you want to dial:
<u>Country/region:</u>	United States (1)
Ar <u>e</u> a code:	300
Phone number:	
Connect using:	СОМ1
	OK Cancel

- 5 In the COM Properties screen, click OK after configuring the following parameters:
 - VT100 terminal emulation
 - Baud rate between 9600 and 230400 bps, the default is 115200 bps.

	No parity,	8 data	bits, 1	l stop	bit,	flow	control	set to	hardware
--	------------	--------	---------	--------	------	------	---------	--------	----------

Bits per second:	115200	*
<u>D</u> ata bits:	8	*
Parity:	None	*
<u>S</u> top bits:	1	*
Elow control:	Hardware	~
	Be	estore Defaults

٠

- **6** The **DTR** LED is on. If it is off, check the HyperTerminal settings and your cable connections.
- 7 Type in AT and press [ENTER]. The terminal will response OK as shown below. You can now use the commands with an "AT" or "at" prefix to configure the TA. Refer to Chapter 5 on page 72 for detailed information on AT command sets.

SIDN TA - HyperTerminal				
02 30	9 1			
AT OK				
Connected 0:00:19 Auto d	letect 115200 8-N-1 SCROLL CAPS NUM Capture Print echo			

3.3 AT Commands

The following tables summarize the AT command sets.

3.3.1 Data Command Sets

The AT data command sets are shown in the following table.

COMMAND	OPTION	DESCRIPTION
A		Answer an incoming call.
C0=n		Select the call type. n=1:HDLC transparent n=3: V.110 n=4: V.120 n=5: X.75 n=8: PPP or MLP (default).
C1=n		This feature is only available when you select PPP or MLP call type (C0=8). Configure maximum throughput by using one B channel or two B channels. n=0: 64 Kbps (one B channel) n=1: 128 Kbps (two B channels)
D <phone number=""> [/<sub-address>] [:<user data="">]</user></sub-address></phone>		Dial an outgoing call. <phone number="">: Enter up to 32 characters, including digits from 0 to 9, *, #, hyphen (-), comma (,) and (','). [','] will be ignored. <sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z". <user data="">: Enter up to 128 printable characters. Note: ATDT, ATDP and ATDW are all same as ATD.</user></sub-address></phone>
DS=n[: <user data="">]</user>		Dial an outgoing call using a pre-stored phone number. n: pre-stored phone number index <user data="">: Enter up to 128 printable characters.</user>
DL		Dial the last-dialed number, including the user data.
E		Enable or disable the command echo function.
	E0	Disable the command echo function.
	E1	Enable the command echo function (default).
Н		Hang up a call.
Ι		Display product information.
	12	Display the model name.
	13	Display the firmware version.
	14	Display the hardware version.
	111	Display the bootcode version.

 Table 14
 AT Data Command Sets Requiring an "AT" Prefix

COMMAND	OPTION	DESCRIPTION
0		Change from escape mode to data mode. +++ is the escape sequence code that is entered in data state to return the ISDN TA to command state. The TA will accept AT commands only while it is in command state.
Q		Enable or disable result code.
	Q0	Enable result code (default).
	Q1	Disable result code.
S0=n		n=0: DIsable the auto answer feature.
		n=1-255: Enable the auto answer feature to answer a call after it rings for n times.
S1		Display read-only last ringing counting for the current incoming call.
S2=n		Set the monitored 3 continuous characters (ESC-Sequence) to enter command mode. $n=0~127$, the default is 43, that is "+++".
S3=n		Set the character value as CR.
		n=0~127, the default is 13.
S4=n		Set the character value as LF. n=0~127, the default is 10.
S5=n		Set the character value as BS (Backspace). n=0~127, the default is 8.
S7=n		The time (in seconds) you wait before connection. n=1-255, the default is 60.
S12=n		Set the "Guard time" for "ESC-Sequence" (in company with "+++"). The unit is 0.1 second. n=0: ignore "Guard time". n=1-255, the default is 10.
S25=n		The DTR transition time. The unit is 0.05 second. n=1-255, the default is 1.
S50=n		Forwarding character in data mode. n=0: no forwarding character (default) n=2: select CR as forwarding character
S51=n		Set how much time the TA is left idle in data mode. The unit is 0.01 second.
S80		Display the read-only cause value of disconnection in the last call.
V		Select result code format.
	V0	Display result code in numeric form.
	V1	Display result code in verbose form (default).
W		Display extended information of "RING" result code.
	W0	No extended information.
	W1	Display phone number of calling site.

	Table 14	AT Data Command Se	ets Requiring an "AT"	Prefix
--	----------	--------------------	-----------------------	--------

COMMAND	OPTION	DESCRIPTION
	W2	Display user data.
	W3	Display phone number of calling site and user data.
Х		Enable/Disable extended result code.
	X0	Basic result code (default).
	X1	Extended result code.
Z		Reload user-defined profile.
&C		DCD signal control.
	&C0	Follow DTR signal.
	&C1	ON only if in communication (default).
&D		DTR (Data Terminal Ready) signal control.
	&D0	Ignore DTR signal, assume DTR is always ON.
	&D1	DTR OFF-ON transition causes dial of an outgoing call using the first pre-stored phone number.
	&D2	DTR OFF causes the modem to hang up (default).
&F		Restore the factory default.
&K		Set flow control mode.
	&K0	Disable flow control.
	&K3	Enable CTS/RTS flow control (default).
	&K4	Enable XON/XOFF flow control.
&S		DSR (Data Set Ready) signal control.
	&S0	Follow DTR signal (default).
	&S1	Follow DCD signal.
&V		Display all data parameters settings.
&W		Save current parameters setting, include &-leading and non- leading parameters, as user defined profile.
%D0=n		Select DTE baud rate when system is powered on.
		n=8: 9600 bps
		n=11: 19200 bps
		n=13. 36400 bps n=16: 57600 bps
		n=17: 115200 bps (default)
		n=21: 230400 bps
%D1=n		Select DTE attribute when system is powered on.
		n=0: 8 data bits, non parity (default)
		n=1: 7 data bits, even parity
%D2=n		Dial an outgoing data call automatically if there is pro-stored
/002-11		dialing number after system is powered on. n=0: do nothing (default)
		n=1: auto-dial after power on

 Table 14
 AT Data Command Sets Requiring an "AT" Prefix

COMMAND	OPTION	DESCRIPTION
%D3=n		The number of retrying to re-dial the outgoing data call once TA failed to establish the data call. n=0: don't re-dial. n=1~10: re-dial for n times until the call is established.
%Id= <phone number> [/<sub- address>]</sub- </phone 		Create a list of the incoming call phone number via the data port. d=0-4. <phone number="">: Enter up to 32 characters, including digits from 0 to 9, *, #, hyphen (-), comma (,) and (','). [','] will be ignored. <sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z".</sub-address></phone>
%M1=n		Enable/Disable PAP to CHAP conversion. n=0: Disable CHAP (default). n=1: Enable CHAP.
%M2=n		Enable/Disable "Dynamic Bandwidth Allocation for Voice Call". n=0: Disable. n=1: Enable (default).
%M3=n		Set threshold of data loading to add one channel, the unit is kbps. n=0-64, the default is 35.
%M4=n		Set a threshold of data loading to drop one channel, the unit is kbps. n=0-64, the default is 25.
%M5=n		Persist time for monitoring threshold to add or drop one channel, the unit is 5 seconds. n=1-255, the default is 24.
%Od= <prefix digits=""></prefix>		Create a list of the outgoing call prefix digits for the data port. d=0-2.
%P0=n		Set incoming call screen mode for the data port. n=0: Accept all incoming calls (default). n=1: Reject all incoming calls. n=2: Only accept the calls from the phone number in the list (see %Id= <phone number=""> [/<sub-address>]). n=3: Only reject the calls from the phone number in the list (see %Id=<phone number=""> [/<sub-address>]).</sub-address></phone></sub-address></phone>
%P1=n		Set outgoing call screen mode for the data port. n=0: No restriction (default). n=1: Prohibit all outgoing calls. n=2: Only accept the calls with the prefix in the pre-defined list (see %Od= <prefix digits="">). n=3: Only reject the calls with the prefix in the pre-defined list (see %Od=<prefix digits="">).</prefix></prefix>

 Table 14
 AT Data Command Sets Requiring an "AT" Prefix

COMMAND	OPTION	DESCRIPTION				
%P2=n		Select CLIP (Calling Line Identification Presentation) or CLIR (Calling Line Identification Restriction).				
		n=0 indicates you select CLIR and the ISDN switch will not pass your phone number to the destination site.				
		n=1 indicates you select CLIP and the ISDN switch will pass your phone number to the destination site.				
%S0= <phone number> [/<sub- address>]</sub- </phone 		Assign a phone number to the data port. <phone number="">: Enter up to 32 characters, including digits from 0 to 9, *, #, hyphen (-), comma (,) and (','). [','] will be ignored.</phone>				
		<sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z".</sub-address>				
-P0=n		Select transmitting frame size for X.75 n=128-2048 bytes, the default is 254.				
-Zd= <phone number> [/<sub-< td=""><td></td><td>Set a pre-stored phone number for the data port. d: 0-4</td></sub-<></phone 		Set a pre-stored phone number for the data port. d: 0-4				
address>j		one number>: Enter up to 32 characters, including digits from 0 to 9, *, #, hyphen (-), comma (,) and (','). [','] will be ignored.				
		<sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z".</sub-address>				

 Table 14
 AT Data Command Sets Requiring an "AT" Prefix

3.3.2 Command Sets for Analog Ports

The AT command sets for the analog port are shown in the following table.

COMMAND	DESCRIPTION
\P0=n	Select CODEC type. Do not change the type until you use the TA in another country. n=0: A-law n=1: µ -law
\P1=n	Select the dialing method for the analog port. n=0 indicates overlap sending, and same system as the traditional telephone (default). n=1 indicates en-block sending, and you must press "#" after the phone number.
\P2=n	Configure the analog port priority. n=0: Both of the analog ports have same priority. n=1: The PHONE 1 port has higher priority. n=2: The PHONE 2 port has higher priority.
١V	Display the settings for all parameters.

 Table 15
 AT Command Sets for the Analog Port

Table 15	AT Comm	and Sets for the Analog Port

COMMAND	DESCRIPTION						
\Zd= <phone number> [/<sub-< td=""><td>Set a pre-stored phone number for the analog port. d: 0-4</td></sub-<></phone 	Set a pre-stored phone number for the analog port. d: 0-4						
address>]	<pre><phone number="">: Enter up to 32 characters, including digits from 0 to 9, *, #, hyphen (-), comma (,) and (','). [','] will be ignored.</phone></pre>						
	<sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z".</sub-address>						
!pA0=n	Set the device type of the analog equipment connected to the analog port.						
	p: enter 1 or 2 to specify the analog port.						
	1=0: indicates a telephone, telephone with caller ID (type 1), G3 fax or modem (default).						
	n=2: indicates a telephone with call waiting caller ID (type 2).						
!pA1=n	p: enter 1 or 2 to specify the analog port.						
	n=0: Disable call waiting.						
	n=1: Enable call waiting (default).						
!pld= <phone number> [/<sub-< td=""><td>Create a list of the phone numbers for incoming call screening on the analog port.</td></sub-<></phone 	Create a list of the phone numbers for incoming call screening on the analog port.						
address>j	p: enter 1 or 2 to specify the analog port. d=0-4.						
	<pre><phone number="">: Enter up to 32 characters, including digits from 0 to 9, *, #, hyphen (-), comma (,) and (','). [','] will be ignored.</phone></pre>						
	<pre><sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z".</sub-address></pre>						
!pOd= <prefix digits=""></prefix>	Create a list of the prefix digits for outgoing call screening on the analog port.						
	p: enter 1 or 2 to specify the analog port.						
	d=0-2						
!pP0=n	Set incoming call screen mode for the analog port.						
	p: enter 1 or 2 to specify the analog port.						
	n=0: Accept all incoming calls (default).						
	n=1: Reject all incoming calls.						
	n=2: Only accept the calls from the phone number in the list (see !pId= <phone number=""> [/<sub-address>]).</sub-address></phone>						
	n=3: Only reject the calls from the phone number in the list (see <code>!pId=<phone number=""> [/<sub-address>]</sub-address></phone></code>).						
!pP1=n	Set outgoing call screen mode for the analog port.						
	p: enter 1 or 2 to specify the analog port.						
	n=0: No restriction (default).						
	n=1: Prohibit all outgoing calls.						
	n=2: Only accept the calls with the prefix in the pre-defined list (see !pOd= <prefix digits="">).</prefix>						
	n=3: Only reject the calls with the prefix in the pre-defined list (see !pOd= <prefix digits="">).</prefix>						
!pP2=n	Select CLIP (Calling Line Identification Presentation) or CLIR (Calling Line Identification Restriction) for the analog port.						
	p: enter 1 or 2 to specify the analog port.						
	n=0 indicates you select CLIR and the ISDN switch will not pass your phone number to the destination site.						
	n=1 indicates you select CLIP and the ISDN switch will pass your phone number to the destination site.						

COMMAND	DESCRIPTION					
!pS= <phone< td=""><td>Assign you phone number to the analog port.</td></phone<>	Assign you phone number to the analog port.					
number> [/ <sub-< td=""><td>p: enter 1 or 2 to specify the analog port.</td></sub-<>	p: enter 1 or 2 to specify the analog port.					
address>j	<pre><phone number="">: Enter up to 32 characters, including digits from 0 to 9, *, #, nyphen (-), comma (,) and (','). [','] will be ignored.</phone></pre>					
	<sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z".</sub-address>					
\$D99	Upgrade firmware.					
\$C0?	Display the charge information of the last call.					
	Use \$c0=0 to clear the information.					
\$C1?	Display the accumulative charge information of the calls.					
	Use \$c0=1 to clear the information.					
\$F99	Reset all parameters to the factory defaults.					
\$P5= <country selection code></country 	Set three-digit country code for caller ID signaling (FSK or DTMF). The default is 216.					

 Table 15
 AT Command Sets for the Analog Port

3.3.3 Debug Command Sets

The AT command sets for debugging are shown in the following table.

Table 16 Debud Command Sets	Table 16	Debug Command Sets
-----------------------------	----------	--------------------

COMMAND	DESCRIPTION
#C0	Clear frame logs.
#Dn	Display frame logs or internal task status. n=80: display frame logs. n=82: display C/I code (layer 1 signal) logs.
L6=n	Set the level of frame logs to Q.921. n=0: disable frame logging. n=1: enable signal frame logging. n=2: enable full frame logging.
L8=n	Set the level of frame logs to V.120, X.75. n=0: disable frame logging. n=1: enable signal frame logging. n=2: enable full frame logging.
L13=n	Set the level of frame logs to PPP. n=0: disable frame logging. n=1: enable signal frame logging. n=2: enable full frame logging.

3.4 Configuring Examples with AT Commands

This section shows you some examples of configuring the TA using AT commands.

3.4.1 Phone Number Assignment

Enter AT ! pS=<phone number> and press [ENTER] to assign the first phone number to the PHONE 1 port.

Enter AT!pS=<phone number> and press [ENTER] to assign the second phone number to the PHONE 2 port.



🏶 ISDN TA - HyperTerminal								
<u> Eile Edit Yiew Call Iransfer Help</u>								
0 🗳 🗑 💈 🕛 🗳 🛙								
AT OK AT!1S=5009001 OK AT!2S=5009002 OK								
								in the second
Connected 0:02:47 Auto detect	19200 8-N-1	SCROLL	CAPS	NUM	Capture	Print echo		

3.4.2 Storing Phone Number

To store a phone number (1234567 for example) on the TA, use the AT-Zd=<phone number> [/<sub-address>] command for data calls and the AT\Zd=<phone number> [/<sub-address>] command for voice calls.





3.4.3 Setting Ports Priority

To set the same priority for both ports, enter $AT \setminus P2=0$ and press [ENTER].

To set the **PHONE 1** port to have the first priority, enter AT\P2=1 and press [ENTER].

To set the **PHONE 2** port to have the first priority, enter $AT \setminus P2=2$ and press [ENTER].

3.4.4 Disabling Call Waiting

A call waiting tone will interfere with data communication. It is highly recommended that you disable the call waiting feature on the port to which a fax or modem is connected.

To disable call waiting on either port, enter AT!1A1=0 or AT!2A1=0 and press [ENTER].

3.4.5 Link Authentication

Refer to Section 2.5.1 on page 32 for more information.

Table 17	PPP AT	commands
----------	--------	----------

AT COMMANDS	DESCRIPTION	
AT%M1=0	Disable PAP to CHAP conversion (default).	
AT%M1=1	Enable PAP to CHAP conversion.	

3.4.6 Call Bumping (Bandwidth Release) for Voice Calls

Refer to Section 2.5.2 on page 34 for more information.

 Table 18
 Call Bumping AT commands

AT COMMANDS	DESCRIPTION	
AT%M2=0	Disable bandwidth release for voice calls.	
AT%M2=1	Enable bandwidth release for voice calls (default).	

3.4.7 Bandwidth On Demand (BOD)

Refer to Section 2.5.3 on page 34 for more information.

AT COMMANDS	DESCRIPTION
AT%M3=n	n= 0-64 (Kbps), the default is 35. Set the maximum threshold. The TA adds the second B-channel when traffic is above the threshold.
AT%M4=n	n= 0-64 (Kbps), the default is 25. Set the minimum threshold. The TA drops the second B-channel when traffic is below the threshold.
AT%M5=n	n= 1-255. The unit is 5-second. The default is 24, that is, 2 minutes. Set the time interval between calculating traffic load.

Note: The high threshold must be greater than the low threshold, otherwise, both of them will be reset to the defaults as shown in the table.

If you set M2=1, M3=0, M4=0 and M5=1, the BOD function will be disabled and the TA establishes fixed multilink PPP data call with both two B-channels and release bandwidth for voice calls.

3.4.8 Incoming Call Screening

Use this function to set restrictions on incoming calls. Refer to Section 3.4.8 on page 64 for more information.

3.4.8.1 Data Call

Use AT%Id=<phone number> [/<sub-address>] to configure a list of incoming call phone numbers (with sub-addresses), then use AT%P0=n to reject or accept the calls with the phone number in the list.

3.4.8.2 Voice Call

Use AT!pId=<phone number> [/<sub-address>] to configure a list of incoming call phone numbers (with sub-addresses), then use AT!pP0=n to reject or accept the calls with the phone number in the list.

3.4.9 Outgoing Call Screening

Use this function to set restrictions on outgoing calls, such as allowing or prohibiting a longdistance or international call. Refer to Section 3.4.9 on page 65 for more information.

3.4.9.1 Data Call

Use AT%Od=<prefix digits> to configure a list of prefix digits for outgoing calls, then use AT%P1=n to prohibit or allow the calls with the prefix digits in the list.

3.4.9.2 Voice Call

Use AT!pOd=<prefix digits> to configure a list of prefix digits for outgoing calls, then use AT!pP0=n to prohibit or allow the calls with the prefix digits in the list.

Figure 21 AT Command Example 3

🗞 ISDN TA - HyperTerminal		J
<u>Elle Edit View Call Iransfer Help</u>		
AT!100=002 OK AT!101=003 OK -		1 March 1
L Connected 0:01:48 Auto detect 19200 8-N-1 SCROLL CAPS NLIM Capture Print echo	<u> </u>	ALL ROOM

3.4.10 Caller ID

Use AT ! pA0=n to select the device type to enable call waiting caller ID when you have a call (see Section 3.4.10 on page 65 for more information).

3.5 Result Codes

A result code is the command response or the connect message to the computer connected to the TA. The format of the result code is dependent on Xn and Vn commands. The result codes are shown in the following table.

NUMERIC FORM	VERBOSE FORM	BASIC (X0)	EXTENDED (X1)
0	ОК	V	V
1	CONNECT	V	
2	RING	V	V
3	NO CARRIER	V	V
4	ERROR	V	V
18	CONNECT 9.6K		V
21	CONNECT 19.2K		V
23	CONNECT 38.4K		V
26	CONNECT 57.6K		V
27	CONNECT 115.2K		V
31	CONNECT 230.4K		V
33	CONNECT 64K		V
34	CONNECT 128K		V

Table 20 Result Codes

CHAPTER 4 Using Tone Commands

You can also use a telephone keypad to configure the TA using tone commands.

4.1 How to Use Tone Commands with a Telephone Keypad

- 1 Make sure your TA hardware is properly connected (refer to the Quick Start Guide).
- 2 Pick up the receiver of the telephone which is connected to the PHONE 1/2 port.
- **3** Issue the *** tone command by dialing the asterisk (*) key three times. You will then hear a configuration tone (refer to Table 21 on page 68 for more information).
- **4** You can now use the tone command(s) to configure the TA. If the command is valid, you will hear a configuration confirmation tone. If the command is not valid, the TA rejects it and generates a busy tone. Hang up the phone and repeat steps 2 and 3 to issue the command(s) again.

4.1.1 Tone and Ring Patterns

The following table describes the tones and ring patterns.

TONE	RING PATTERN	
Dial Tone	Depend on your ISDN provider.	
Local Dial Tone	Listing on.	
Busy Tone	0.5 Sec On, 0.5 Sec Off, Repeat.	
Call Waiting Tone	0.125 Sec On, 0.125 Sec Off, 0.125 Sec On, 0.625 Sec Off, Repeat.	
Configuration Tone	0.125 Sec On, 0.125 Sec Off, Repeat	
Configuration Confirm Tone	1.5 Sec On, 0.5 Sec Off, Repeat	
Sub Command Tone	0.125 Sec On, 0.125 Sec Off, Repeat	
Normal Ring Patter	1 Sec On, 2 Sec Off, Repeat	
	This ring pattern varies depending on the country code you select (see Table 9 on page 44).	
Remind Ring	0.5 Sec On, 0.5 Sec Off, Repeat	

	Table 2	21 To	one and	l Ring	Patterns
--	---------	-------	---------	--------	----------

4.2 Tone Commands

The tone command sets are shown in the following table. Refer to the corresponding AT commands in Chapter 5 on page 72 for more information.

TONE COMMAND	AT COMMAND	DESCRIPTION
991*99#	\$F99	Reset all parameters to the factory defaults.
015* <country code="">#</country>	\$P5	Set three-digit country code for caller ID signaling (FSK or DTMF). The default is 216.
120*n#	\P0	Select CODEC type. Do not change the type until you use the TA in another country. n=0: A-law n=1: µ -law
121*n#	\P1	Select the dialing method for the analog port. n=0 indicates overlap sending, and same system as the traditional telephone (default). n=1 indicates en-block sending, and you must press "#" after the phone number.
122*n#	\P2	Configure the analog port priority. n=0: Both of the analog ports have same priority. n=1: The PHONE 1 port has higher priority. n=2: The PHONE 2 port has higher priority.
13d* <phone number=""> [*<sub-address>]#</sub-address></phone>	\Zd	Set a pre-stored phone number for the analog port. d=0-4: phone number index. <phone number="">: Enter up to 32 characters, including digits from 0 to 9, *, #, hyphen (-), comma (,) and (','). [','] will be ignored. <sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z".</sub-address></phone>
210*n#	!pA0	n=0: Disable the Type 2 caller ID service on the current analog port. n=2: Enable the Type 2 caller ID service on the current analog port.
211*n#	!pA1	n=0: Disable call waiting. n=1: Enable call waiting (default).
220*n#	!pP0	Set incoming call screen mode for the analog port. n=0: Accept all incoming calls (default). n=1: Reject all incoming calls. n=2: Only accept the calls from the phone number in the list (see 32d* <phone number=""> [*<sub-address>]#). n=3: Only reject the calls from the phone number in the list (see 32d*<phone number=""> [*<sub-address>]#).</sub-address></phone></sub-address></phone>

TONE COMMAND	AT COMMAND	DESCRIPTION
221*n#	!pP1	Set outgoing call screen mode for the analog port.
		n=0: No restriction (default).
		n=1: Prohibit all outgoing calls.
		n=2: Only accept the calls with the prefix in the pre-defined list (see 35d* <prefix digits="">#).</prefix>
		n=3: Only reject the calls with the prefix in the pre-defined list (see 35d* <prefix digits="">#).</prefix>
222*n#	!pP2	Select CLIP (Calling Line Identification Presentation) or CLIR (Calling Line Identification Restriction) for the analog port.
		n=0 indicates you select CLIR and the ISDN switch will not pass your phone number to the destination site.
		n=1 indicates you select CLIP and the ISDN switch will pass your phone number to the destination site.
310* <phone number=""></phone>	!pS	Assign your phone number to the analog port.
[* <sub-address>]#</sub-address>		<pre><phone number="">: Enter up to 32 characters, including digits from 0 to 9, *, #, hyphen (-), comma (,) and (','). [','] will be ignored.</phone></pre>
		<pre><sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z".</sub-address></pre>
32d* <phone number=""> [*<sub-address>]#</sub-address></phone>	!pld	Create a list of the incoming call phone number via the analog port.
		d=0-4: phone number index.
		<pre><phone number="">: Enter up to 32 characters, including digits from 0 to 9, *, #, hyphen (-), comma (,) and (','). [','] will be ignored.</phone></pre>
		<sub-address>: Enter up to 20 ASCII characters ranging from "0-9"," a-z" and "A-Z".</sub-address>
35d* <prefix digits=""> #</prefix>	!pOd	Create a list of the outgoing call prefix digits via the analog port. d=0-2

Table 22Tone Commands

4.3 Configuring Example with Tone Commands

This section shows you some examples of configuring the TA using Tone commands. Refer to Chapter 2 on page 24 for more information about these TA features.

4.3.1 Phone Number Assignment

Use the ***310*<phone number>[*<sub-address>] # tone command to assign a phone number to the **PHONE** port to which the telephone is attached. For example, dial ***130*5009001# to assign 5009001 to the analog port.

4.3.2 Storing Phone Number

To store a phone number on the TA, use the 13d*<phone number> [*<subaddress>] # command for voice calls. For example, dial ***130*1234567# to store 1234567 on the TA.The phone number index is 0. See Section 2.12.1 on page 46 on how to place a voice call.

4.3.3 Setting Ports Priority

To set the same priority for both two ports, use the ***122*0# command.

To set the **PHONE 1** port to have the first priority, use the ***122*1# command.

To set the **PHONE 2** port to have the first priority, use the ***122*2# command.

4.3.4 Disabling Call Waiting

To disable call waiting on either port, use the ***211*0# command.

4.3.5 Incoming Call Screening

Use this function to set restrictions on incoming calls.

4.3.5.1 Voice Call

Use 32d*<phone number> [*<sub-address>] # to configure a list of incoming call phone numbers (with sub-addresses), ***320*1234567# for example, then use <math>220*n# to reject or accept the calls with the phone number in the list.

4.3.6 Outgoing Call Screening

Use this function to set restrictions on outgoing calls, such as allowing or prohibiting a longdistance or international call.

4.3.6.1 Voice Call

Use 35d*<prefix digits># to configure a list of prefix digits for outgoing call, ***350*080# for example, then use 221*n# to prohibit or allow the calls with the prefix digits in the list.

4.3.7 Caller ID

Use $210 \times n\#$ to select the device type to enable call waiting caller ID when you have a call (see Section 3.4.10 on page 65 for more information).
CHAPTER 5 Firmware Upgrade

You can upgrade TA firmware using the utility or by terminal emulation program such as HyperTerminal in Windows. To obtain the latest firmware version, please go to ZyXEL's website at http://www.zyxel.com. Only use firmware for your TA's specific model.

5.1 Upgrading Using the Utility

- 1 Follow the steps in Section 2.1 on page 24 to open the utility.
- 2 Click the Upgrade icon on the navigation panel.

Figure 22 Utility: Navigation Panel

ZyXEL Omni	.Net Lite	ISDN TA U	tility Ver 1.2	Firmw	are Ver 1.0		
C Load Default	P Load	Save	Submit to TA	Upgrade V) <u>メ</u> Exit		
E 🌽 TA	item		l ocal	Directory Nut	nher		

3 When the **Open** screen displays, browse to find the file you want to upload and click **Open**.

Figure 23 Utility Firmware Upgrade: Open a File

Open				? 🔀
Look jn: 🔀) Firmware	•	🗢 🔁	* 🎟 •
Continues Contin	Lite Firmware Release Noto teE2V1.0	e V1.O		
File <u>n</u> ame:	OmniNetLiteE2V1			<u>O</u> pen
Files of type:			-	Cancel

4 A window display asking you to confirm that you want to upload firmware. Click OK.

Figure 24 Utility Firmware Upgrade: Confirm

Upgrade firmwa	re! Are you sure?
OK	Cancel

5 The following screen displays while firmware upload is in progress.

Note: Do not interrupt the file transfer process as this may PERMANENTLY DAMAGE your TA.

Figure 25 Utility Firmware Upgrade: In Progress

Firmware u	pgrade progress status	
	19%	

6 After a successful upload, the system will reboot.

Figure 26 Utility Firmware Upgrade: Restart

😰 Rebooting Status	
Rebooting TA	27

5.2 Upgrading Using HyperTerminal Example

1 Follow the steps in Section 3.1 on page 52 to start HyperTerminal in Windows.

2 Type AT\$D99 and press [ENTER]. A message "Remote Firmware Upgrade Start..." shows in the next screen.

🗞 ISDN TA - HyperTerminal	
Eile Edit View Gall Iransfer Help	
AT OK AT\$D099 Remote Firmware Upgrade Start CCCC	
Connected 0:00:52 Auto detect 19200 8-N-1 SCROLL CAPS NUM Capture Print echo	

- **3** Be sure you have the new firmware file on hand. You must decompress compressed (.zip) files before you can upload them.
- 4 Click Transfer, then Send File to display the Send File screen.



5 Choose the **1K Xmodem** protocol. Type the firmware file's location, or click **Browse** to search for it. Then click **Send**.

Send File	? 🛛
Folder: D:\OmniNet Lite\Fw\Omni.Net Lite <u>F</u> ilename:	
D:\OmniNet Lite\Fw\Omni.Net Lite FirmwareV1.0	Browse
Protocol:	
1K Xmodem	*
<u>S</u> end <u>C</u> lose	Cancel

6 Firmware upload is in progress; do not turn off the TA or disconnect the RS-232 cable.

senaing.	D. YOMPINE	Literrevolution	Lite Finnware	V1.01 VFIIMWale (UMPIINELL
Packet:	297	Error checking:	CRC	
Retries:	0	Total retries:	0	
Last error:				
File:				37K of 322K
Elapsed:	00:00:51	Remaining:	00:06:36	Throughput: 735 cps

7 After the firmware upload process has completed, the TA automatically restarts. You can check your new firmware version by using the ATI3 command.



CHAPTER 6 Troubleshooting

This chapter covers potential problems and possible remedies. After each problem description, some instructions are provided to help you to diagnose and to solve the problem. Please see our included disk for further information.

PROBLEM	CORRECTIVE ACTION		
None of the LEDs are on.	Make sure the TA's power adaptor is connected to the TA and an appropriate power source. Check that the TA and the power source are both turned on.		
	Make sure your computer or terminal is turned on.		
	Check all cable connections.		
	Turn the TA off and on. If the error persists, you may have a hardware problem. In this case, you should contact your local vendor		
I cannot connect to the Internet.	Click Start, Control Panel, double-click Phone and Modem Options (or Modems). If Omni.Net Lite TA is not present, install the driver.		
	Make sure you entered the correct phone number, user name and password.		
	Make sure the ISDN port is properly connected to the ISDN wall jack with a RJ-45 cable.		
	Connect a phone into the PHONE 1 or PHONE 2 port and see if you get a dial tone.		
	Carefully follow the instructions to uninstall and reinstall the software driver.		
	Restart your computer		
Windows does not auto-detect the RS- 232 connection to the TA.	Make sure the TA is turned on and connected to a serial port on the computer.		
	Perform a hardware scan by clicking Start , Settings , Control Panel and double-clicking Add/Remove Hardware . (Steps may vary depending on the version of Windows). Follow the on-screen instructions to search for the modem and install the driver.		
	Check for possible hardware conflicts. In Windows, click Start , Settings , Control Panel , System , Hardware and then click Device Manager . Verify the status of the modem under Network Adapters . (Steps may vary depending on your version of Windows).		
	Connect the TA to another computer. If the error persists, you may have a hardware problem. In this case, you should contact your local vendor.		
I need two B-channels	By default, the TA uses dynamic 128K MLP.		
for Internet access, but only one B-channel is active.	Use the utility to enable fixed MP to have the TA always transmit data at 128 Kbps or configure the minimum and maximum thresholds for adding or adding the second B-channel according to traffic load (see Section 2.5 on page 32).		
The TA does not	Make sure you have subscribed to the call waiting service first.		
release bandwidth for incoming voice calls.	Check if the call bumping feature is activated.		

Table 23 Troubleshooting

Index

Numerics

110V AC 4 230V AC 4

A

AC 4 Accessories 4 Airflow 4 American Wire Gauge 4 AT Commands 55 AT Commands Overview 52 AWG 4

В

BACP 29 BACP and BAP 29 Bandwidth On Demand 19, 34 Bandwidth On Demand (BOD) 64 Bandwidth Release 34, 64 BAP 29 Basement 4 Basic Rate Interface 18 BOD 34, 64 BRI 18 busy tone 68

С

Cables, Connecting 4 Call Bumping 18, 34, 64 Call Forwarding 19 Call Hold 48 Call Transfer 19, 50 Call Waiting 19, 46, 49 Caller ID 44 Caller Identification 44 Calling Line Identity Presentation 46 Calling Line Identity Restriction 46 Challenge Handshake Authentication Protocol 32 CHAP 32 Charge Information 42 Clear to Send 21 CLIP 46 CLIR 46 configuration confirmation tone 68 Configuration Methods 21 configuration tone 68 Connecting Cables 4 Copyright 2 Corrosive Liquids 4 Country Code 27 Covers 4 CTS 21 CTS/RTS Handshake 36 Customer Support 6

D

Damage 4 Dampness 4 Danger 4 Data Carrier Detect 38 Data Set Ready 21 Data Terminal Equipment 21 Data Terminal Ready 21 DCD 38 Denmark, Contact Information 6 Device Type 46 Dialing Method 45 Directory Number 28 Disabling Call Waiting 49 DN 28 DSR 21, 38 DTE 21 DTMF 44 DTR 21, 38, 57 Dust 4

Ε

Electric Shock 4 Electrical Pipes 4 Electrocution 4 En-Bloc Sending 45 Europe 4 Exposure 4

F

FCC 3 Finland, Contact Information 6 Firmware Upgrade 19, 72 Flow Control 36 France, Contact Information 6 FSK 44

G

Gas Pipes 4 Germany, Contact Information 6 getting started 18

Η

hardware connections 20 Hardware Flow Control 36 HDLC 30 High Voltage Points 4 High-Level Data Link Control 30 How to Use Tone Commands with a Telephone Keypad 68 HyperTerminal 72 Hyperterminal 52

I

Incoming Call Screening 38, 64 Incoming Voice Call Scenarios 47 Intercom 47 Introduction to PPP 28 ISDN Interfaces 19 ISDN Protocols 30

L

Lightning 4 Link Authentication 32 Liquids, Corrosive 4

Μ

MLP 18, 29 MLPPP 18 MP 18, 29 MSN 43, 47 Multilink PPP 29 Multiple Subscriber Number 43

Ν

North America 4 North America Contact Information 6 Norway, Contact Information 6 NT-1 19

0

Off-hook Caller ID Transmission 44 On-hook Caller ID Transmission 44 Opening 4 Outgoing Call Screening 39, 65 Overlap Sending 45

Ρ

PAP **32** PAP and CHAP Security Password Authentication Protocol Phone Number Assignment Phone Number Assignment Example Pipes 4 Point-To-Point Protocol 28 Pool 4 Ports Priority 46, 47, 63 Power Adaptor 4 Power Cord 4 Power Outlet 4 Power Supply 4 Power Supply, repair 4 PPP 28 PPP Multilink 18 PPP/MLP 30 Pre-stored Numbers 40

Q

Qualified Service Personnel 4 Quick Start Guide 16, 20, 24, 52, 68

R

Regular Mail 6 Related Documentation 16 Removing 4 Repair 4 Request to Send 21 Result Codes 57, 66 Risk 4 Risks 4 RTS 21 RTS Threshold 36 RTS/CTS 36

S

S/T interface 19 Service 4, 5 Service Personnel 4 Shock, Electric 4 Software Flow Control 36 Spain, Contact Information 6 Storing Phone Number 62 Subaddress 43 Supplementary Services 19, 46, 47 Supplementary Voice Features Supply Voltage support CD Support E-mail Sweden, Contact Information Swimming Pool Switch Type syntax conventions System **27**

Т

Telecommunication Line Cord. 4 Telephone 6 The Flash Key 48 Three-Way Conference 51 Thunderstorm 4 Tone and Ring Patterns 68 Tone Commands 69

U

U interface 20 Using the Utility 24

V

V.110 **30** V.24 Signal Vendor **4** Ventilation Slots Voltage Supply Voltage, High

W

Wall Mount 4 Warnings 4 Water 4 Water Pipes 4 Web Site 6 Wet Basement 4 Worldwide Contact Information 6

Χ

X.75 **30** XON/XOFF **36**

Ζ

ZyXEL Limited Warranty Note 5