



Voip Setup Guide for Dynalink (RTA1046VW)

1. Open up your Web browser (EG: internet explorer)
2. Put in the following IP address into the address bar:
192.168.1.1
3. Log into the modem using the following:
 - i. Username = admin
 - ii. Password = admin
4. Click on voice:

The screenshot shows the Dynalink web interface. At the top, there is a navigation bar with the Dynalink logo and several menu items: Quick Start, Status, Advanced, Wireless, Voice, and Management. The 'Voice' menu item is highlighted with a red rectangular box. Below the navigation bar, there is a 'Language: English' dropdown menu. The main content area is titled 'Connect to Internet' and 'Quick Setup'. It displays the status of the DSL router as 'connected' to Broadband. A table shows the current connection status:

Current Connection Status:	
Online Time	3 days 4 hours 28 mins 26 secs
Data Transmitted	
Data Received	

Below the table, there is a 'Disconnect' button and a warning message: 'Clicking "Disconnect" will close down your Broadband connection. This will affect all computers currently connected.'

At the bottom left of the page, the following information is displayed:

Firmware: 3.63u
ADSL2+: A2pB023k.d20h
Wireless: 3.131.35.6

5. Change the “**Interface name**” to “**PPP_8_35_1-pppoe_8_35**”
6. Change “**SIP Transport Protocol**” to “**UDP**”
7. Change “**Voip service provider**” to “**Other**”
8. Please input for the following:
 - i. SIP Register Address = **sip1.exetel.com.au**
 - ii. SIP Outbound Proxy = **sip1.exetel.com.au**
 - iii. SIP Proxy Server Address = **sip1.exetel.com.au**
9. For all ports please input: “5060”
10. Please input for the following (phone1) :
 - i. **Phone Number = voip number** (Sent to you via email)
 - ii. **Call Name = voip number** (Sent to you via email)
 - iii. **User Name = voip number** (Sent to you via email)
11. Please input your password for the voip account (Sent to you via email)
12. Press on the **apply** button
13. If it registers correctly the following message would appear:
Phone 1: VOIP Mode (You can make Voip phone calls through this port now)

- SIP Basic
- SIP Extension
- Audio Codec
- PSTN Control
- Call Log

SIP Basic Settings

Phone 1: **VOIP mode** **13.**
 (You can make VOIP phone calls through this port now.)

Phone 2: **PSTN mode**
 (The registration is failed. You can not make VOIP phone calls through this port now.)

Interface: pppoe_8_35_1 **5.**

SIP Transport Protocol: UDP **6.**

VoIP Service Provider: Others **7.**

SIP Registrar Address: sip1.exetel.com.au

SIP Outbound Proxy: **8.** sip1.exetel.com.au

SIP Proxy Server Address: sip1.exetel.com.au

Port: 5060 **9.**

Port: 5060

Port: 5060

Port: 5060

Registration Information

Phone	Phone Number	Caller Name	User Name	Password
1	02	02 10.	02	***** 11.
2				

Apply Cancel

12.