

PANASONIC KX-UT SERIES SIP TERMINALS FEATURE SHEET

The Panasonic UT range of SIP telephony terminals enhance personal communications through excellent HD quality audio on every phone, combined with easy access to powerful supportive features and applications. The range - from standard phones, office key-sets, executive terminals and touch-screen Smart Desk application phones - addresses all requirements. Panasonic's reputation for design, quality, reliability and care for the environment, ensures an exceptional user experience wherever the terminals are deployed - as part of a "cloud based" service or with an IP PBX - in a business environment or in the home.







KEY FEATURES

High Definition "HD" Audio

The KX-UT series of SIP terminals offers 'best in class' audio quality, meaning fewer repeated conversations and misheard calls. Offering Wideband High Definition Audio as standard across the range, the UT series offer G.722, G.711, G.726 and G.729a Codecs. Coupled with Enhanced Echo Cancellation and an Expanded Acoustic Chamber, the UT series of SIP terminals offers a superior audio experience to users over handset, speakerphone and optional headsets. The entire range features wideband compliant, hearing aid compatible handsets, and built-in, high quality speaker and microphone.

LCD Displays

Large, clear LCD displays with intuitive User Interface offer fast access to phonebooks and features.

Electric Hook Switch

A built in Electronic Hook Switch (Plantronics compliant) port allows the KX-UT133 and KX-UT136 SIP Terminals to have access to the Plantronics range of DECT enabled headsets. This offers a range of portability and comfort as frequent users are able to move around freely, without being tied down by handsets.

ECO Friendly

Low power consumption, combined with an advanced ECO standby mode means lower energy costs. Consumption can be less than 1 watt in ECO mode! (UT 1xx series).

Plug and Play Configuration

Extensive provisioning options, including automatic configuration via a provisioning server, means a smaller administration overhead, saving time and money. The range is certified compatible with Digium Asterisk and Broadsoft Broadworks, ensuring excellent compatibility with leading soft switch suppliers.









						Work
	KX-UT113	KX-UT123	KX-UT133	KX-UT136	KX-UT248*	KX-UT670*
DISPLAY						
LCD Display	Monochrome Graphical	Monochrome Graphical	Monochrome Graphical	Monochrome Graphical	Greyscale Graphical	Colour 262k
LCD Size		242 x 55 pixels – 3 lines		242 x 109 pix. – 6 lines	4.4 inch	7 inch touch
LCD Contrast	6 levels					No
LCD Backlight	None		On/Auto/Off			15/Auto 720p
HD Video	No No					
INSTALL OPTIONS						
Desk mount tilt	No.		Yes – 2 positions			10/ 4/0/ (.: 1)
Wall mount Power adaptor	KX-A43Z (KX-A432 (optional) KX-A433 (optional) Option - KX-A239		KX-A433 (optional)	Option -	KX-A434 (optional)
NETWORKING		υμασι - ι	NA-AZJ7		υμιοιι -	NA-N4ZZ
Ethernet Ports	1 10/100 2 10/100				0 10/1	100/1000
Power over Ethernet (PoE)	1 – 10/100 2 – 10/100 2 – 10 Yes					100/1000
Bluetooth	No Yes					Optional
AUDIO FEATURES						T T
Handset, Speaker, Headset Volume	8 levels (includes echo cancellation and distortion prevention)					15 levels
Ringtones	32					
Ringer Volume	6 levels + Off					7 levels + Off
Headset Port Electronic Hook						
Switch Control Port	wone 5.5 mm Prantronics compatible					
Audio Codec	G.711, G.722, G.726, G.729a G.711, G.722, G.729a					
KEYS						
Soft Keys			Yes - 4			Yes - screen
Programmable Keys	0		24		24 (3 pages of 8 keys)	24 (4 pages of 6 keys)
0-9, *, # keys	Yes Yes					Yes - screen keys
Navigator & Cancel Key			Yes			On screen + menu keys
SOFTWARE FEATURES	100		500	w.e.		1 2 40
Phone Book (Entries)	100 – each with 5 numbers		500 - each with 5 numbers 30 incoming calls + 30 outgoing calls			Depends on memory *2
Call Log Entries	None 30 incoming calls + 30 outgoing calls 3 parties (within terminal — multi-party dependent on server)					
Conferencing XML	3 parties (within terminal — multi-party dependent on server) XML application support from Q1 2012					
Music on Hold	Supplied by Host Service (PBX / SIP Server)					
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IP FEATURES				,		
SIP Accounts SIP Compatibility	2			4 oriek Proodcoft Panacanic SID DD		6
IP Version	RFC 3261 Standard SIP Server, Asterisk, Broadsoft, Panasonic SIP PBX IPv4					
DHCP Client	Yes					
DNS	Yes					
HTTP	Yes					
HTTPS	Yes					
SNTP Client VLAN (802.1q)	Yes Yes					
QoS (DiffServ)	TES Yes					
802.1x	No Yes					es
CONFIGURATION						
Plug & Play Configuration			Server based configuration. TR	-069, Panasonic Redirect Server		
Manual Configuration	Internal web Configurator, Local (LCD based) network configuration					
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^{*}Preliminary Specifications



^{*2} up to 500 connected to NS1000