

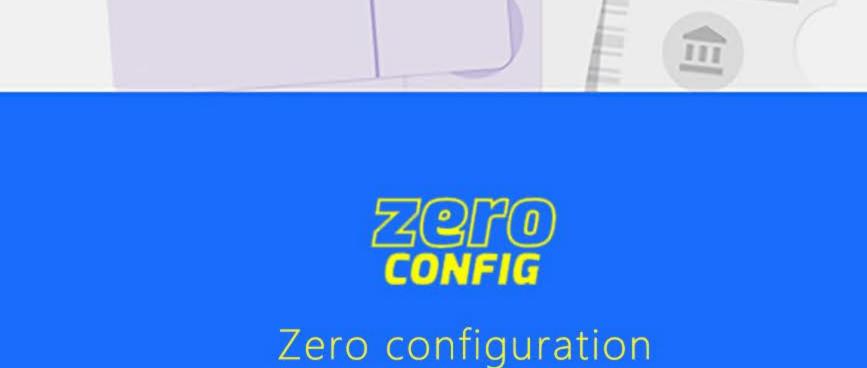


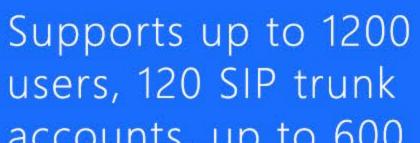
KNTD-1200

The KNTD-1200 IP PBX appliance is designed to bring leading edge voice, video, data,

UC FEATURES FOR UP TO 1200 USERS

and ability features to enterprises, small and medium businesses, retail and residential environments in an easy-to-manage fashion. This enterprise-grade on premise IP PBX supports E1, T1 and J1 networks and offers scalability by supporting up to 1200 users. This secure and reliable IP PBX delivers unifed communication features at an unprecedentedprice point it hout any licensing fees, costs-per-feature, or recurring fees. Host Standby N+1 redundancy System design





MIX 1200 users

accounts, up to 600 concurrent calls



Built-in call recording

server; recordings

accessed via web

user interface



HTTPS encryption

Supports call queue

volume management

for efficient call



to a 5-level IVR

Supports up

efficiently handle incoming calls

Multi-language

auto-attendant to



Integrated

LDAP and XML

Bexible dial plan

phonebooks,

HARDWARE INTERFACE

The third Gigabit port for the hot backup cluster

and integrated POE (IEEE 802.3at-2009)

Support (user configuration)

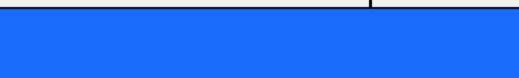
KNTD-1200 IPPBX Enterprise - class unified communication system

Using dual Gigabit interface (10/100 / 1000M Ethernet adaptive)



using the H.264

Supports any SIP video endpoint that



Network Interface

NAT router

Model

Host Standby N+1 redundancy System design

NAT router	Support (user configuration)			
Reset switch	Support, Web configuration			
AUDIO / VIDEO CAPABILITY				
Speech algorithm	Support fax tone detection and automatic switching G.711, 7.29, 7.23, 7.26 encoding			
Voice	G.711 A-law / U-law, G.722, G.723.1, G.726, G.729A / B, AAL2-G.726-32			
Video coding	H.264, H.263, H263 +			
QoS	Layer 3 QoS, Layer 2 QoS			
SIGNALING PROTOCOL				
DTMF detection method Support three, In Audio, RFC2833, SIP INFO				
Digital signal	PRI, SS7, MFC / R2			
Terminal management and automatic deployment capabilities	Support for SIP terminal plug-and-play auto-deployment (including DHCP Option 66 / multicast SIP SUBSCRIBE / MDNS Automatic detection and configuration), support TFTP / HTTP / HTTPS upgrade, local upgrade			
Network protocol	TCP / UDP / IP, RTP / RTCP, ICMP, ARP, DNS, DDNS, DHCP, NTP, TFTP, SSH, HTTP / HTTPS, SIP (RFC3261), STUN, SRTP, TLS / SIP, LDAP			
PSTN	Support the PSTN line intelligent detection and automatic matching			

Digital signal	PRI, SS7, MFC / R2			
Terminal management and automatic deployment capabilities	Support for SIP terminal plug-and-play auto-deployment (including DHCP Option 66 / multicast SIP SUBSCRIBE / MDNS Automatic detection and configuration), support TFTP / HTTP / HTTPS upgrade, local upgrade			
Network protocol	TCP / UDP / IP, RTP / RTCP, ICMP, ARP, DNS, DDNS, DHCP, NTP, TFTP, SSH, HTTP / HTTPS, SIP (RFC3261), STUN, SRTP, TLS / SIP, LDAP			
PSTN disconnection method	Support the PSTN line intelligent detection and automatic matching function, support five kinds of signaling detection methods (1, Busy Tone detection, 2, CPT detection, 3, polarity reversal detection, 4, Hook Flash Timing detection, 5, Loop Current Disconnect detection)			
SAFETY				
Media encryption	SRTP, TLS, HTTPS, SSH			
Advanced defense	Fail2ban, Alert events, whitelists, blacklists, encrypted access controls			
PHYSICAL CHARACTERISTICS				
Global universal power supply	Input: 100 ~ ~ 240VAC, 50 / 60Hz;			
weight	Equipment weight: 2.165 Kg; Package weight: 3.012 Kg			
Use the environment	Operating environment: 32 - 113°F / 0 ~ 45°C, humidity 10 - 90% (No condensation) Storage environment: 14 - 140°F / - 10 ~ ~ 60°C, humidity 10 - 90% (No condensation)			
size	440mm (L) x 185mm (W) x 44mm (H) IU standard			
installation	Desktop or cabinet installation			
	ADDITIONAL FEATURES			
Multi-language support	Web user interface supports English / Simplified Chinese, customizable IVR / voice prompts in English, Chinese, english English			
Caller ID detection	ETSI-FSK, ETSI-DTMF			
Polarity reversal/ Wink	Support, option on / off control call setup and termination			
Call Center	Supports multiple call queues and queue queuing sounds			
Customize Auto Attendant	Supports Layer 3 IVR (Interactive Voice Response)			
Concurrent call capability	Up to 5,000 SIP terminals can be registered, up to 500 concurrent calls			

KNTECH IP PABX MODEL NUMBER

SIMULTANEOUS CALLS

Pickup Group, blacklist, call / intercom, etc.

Supports up to 10 voice conference rooms, up to 100 voice conference

Call Park, Call Forward, Call Forward, Do Not Disturb, DISA, ring Group,

REGISTER USER	SIMULTANEOUS CALLS	MODEL NUMBER		
50 line	15 line	KNTD-50		
100 line	30 line	KNTD-100		
300 line	90 line	KNTD-300		
500 line	150 line	KNTD-500		
800 line	400 line	KNTD-800		
1000 line	300 line	KNTD-01K		
3000 line	900 line	KNTD-03K		
5000 line	1500 line	KNTD-05K		
HONGKONG KOON TECHNOLOGY LIMITED				

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members

Meeting room

Call function

capacity

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