



KNTD-800

The KNTD-800 IP PBX appliance is designed to bring leading edge voice, video, data,

UC FEATURES FOR UP TO 800 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 800 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. Hcde Standby N +1 redundancy System design



Supports up

to a 5-level IVR

MIX 800 users

Supports up to 800

users, 80 SIP trunk

concurrent calls

accounts, up to 400

Response)

(Interactive Voice



provisioning of

KNTECH SIP

End points

Built-in call recording

server; recordings

accessed via web

user interface



Strongest-possible

security protection

Supports call queue for efficient call volume management



Multi-language

auto-attendant to

efficiently handle

incoming calls

Hcde Standby N +1 redundancy System design Model KNTD-800 IPPBX Enterprise - class unified communication system

Network Interface

Network protocol

Advanced defense

Global universal

Multi-language

Caller ID detection

support

capability

capacity

Meeting room

Call function

power supply

NAT router

Reset switch



Integrated

LDAP and XML

Bexible dial plan

phonebooks,

HARDWARE INTERFACE

and integrated POE (IEEE 802.3at-2009)

Support (user configuration)

Support, Web configuration

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



Supports any SIP

using the H.264

video endpoint that

The third Gigabit port for the hot backup cluster

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 me	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			

TCP / UDP / IP, RTP / RTCP, ICMP, ARP, DNS, DDNS, DHCP, NTP, TFTP,

SSH, HTTP / HTTPS, SIP (RFC3261), STUN, SRTP, TLS / SIP, LDAP

Support the PSTN line intelligent detection and automatic matching

Fail2ban, Alert events, whitelists, blacklists, encrypted access controls

Web user interface supports English / Simplified Chinese, customizable

Up to 5,000 SIP terminals can be registered, up to 500 concurrent calls

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

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

IVR / voice prompts in English, Chinese, english English

PSTN function, support five kinds of signaling detection methods (1, Busy disconnection Tone detection, 2, CPT detection, 3, polarity reversal detection, 4, method Hook Flash Timing detection, 5, Loop Current Disconnect detection) SAFETY Media encryption SRTP, TLS, HTTPS, SSH

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		

PHYSICAL CHARACTERISTICS

Input: 100 ~ ~ 240VAC, 50 / 60Hz;

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	Up to E 000 CID terminals can be registered up to E00 concurrent call

Pickup Group, blacklist, call / intercom, etc.

ETSI-FSK, ETSI-DTMF

members

KNTECH IP PABX MODEL NUMBEF
SIMULTANEOUS CALLS

	SIIVIOLTAINLOUS CALLS	
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

1001 T3 Yifang Center Baoan Shenzhen China

HONGKONG KOON TECHNOLOGY LIMITED

Tel: 0086755 29116322 Fax:0086755 29790278 Email: marketing@koontech.com www.koontech.com