TELESIS mL series Hybrid IP-PBX

TELESIS - Leading-Edge IP Telephony System, is a Hybrid IP PBX Business Phone Systems that are capable of interworking between all its IP and TDM users. It has a new approach to telephony technology that is CAU -Cluster of Autonomous Units, providing a Fully Features IP-PBX with built-in LAN Port and 112 SIP Licenses (Common for SIP Trunk and SIP Extensions)



Hybrid IP PBX Business Phone Systems

- Hardware Platform
 - 1 Cabinet having 4 slots for 64 TDM Ports
 - 2 Cabinets combo having 8 slots for 128 TDM Ports
 - External 48 V DC Power Adaptor
 - Interface
 - SIP Trunk Lines
 - ISDN PRI Line
 - Analog CO Lines with CLI
 - SIP Extensions
 - Analog Extensions
 - Digital extensions
 - Software
 - Built-in 112 SIP Port Licenses (common to SIP Trunk and SIP Extensions



Main Cabinet

Expansion Cabinet

CPU 1 2 3 4









TELESIS Hardware Configuration

- Each Cabinet has 4 Slots (64 TDM Ports)
- 2 Cabinet Stacking with Flat Cable ,provides 128 TDM Ports
- Built-in 112 SIP Port Licenses (Common for SIP Trunk & SIP Extensions)
- **Analog extension Cards** 8 Port and 16 Ports
- **Analog Trunk Cards** 4 Port and 8 Ports
- **Hybrid Cards** 4 CO x 8 Analog Extensions / 4 CO x 2 Digital x 6 Analog
- **Digital Cards** 8 Port and 15 Port Digital Cards
- ISDN PRI Card Mounted on CPU , No slot Required, Max 1 ISDN PRI Card
- Max Capacity: 1 PRI / 4 CO / 120 Analog / 112 SIP Licenses / 16 xSIP
 - (Connect Upto 127 Systems in clusters on IP network and get upto 30,000 ports)



DETAILS OF FEATURES



And many other built-in features...

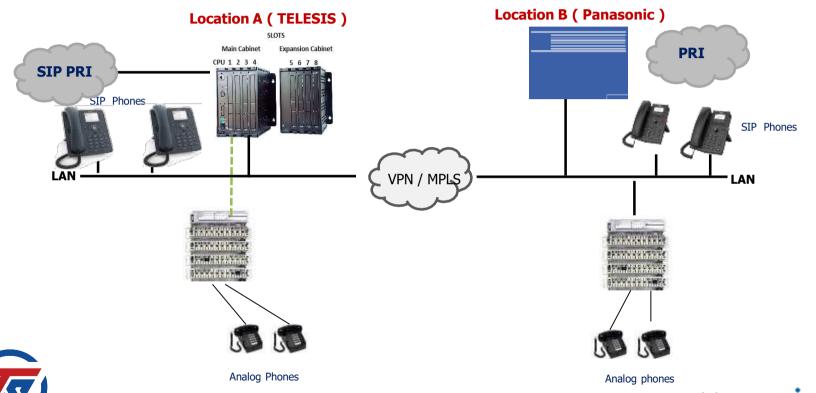
- Voicemail & Voice to email
 - Unlimited number of mailboxes with 100 Hours of Voice Mail
- Hotel Features & Flexible Numbering
 - including direct PMS Interface
- 100 Hours Call Recording
- Hunt Groups
- Multiple Level IVR / Auto Attendant

- Multiple SIP Trunk Interface
 - Use Dual WAN Router and connect Airtel & TATA on same PBX
- Multi-Party Conference
- CDR
 - use TELESIS Tool to search and download Call Details
 - BLF (Busy Lamp Field) on SIP Phones , like SNOM, intele T 790, etc





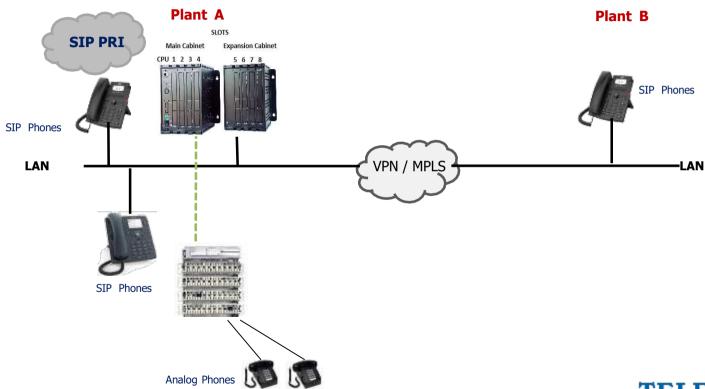
Connect 2 PBX system on Intra Branch Network



DETAILS OF SOLUTIONS



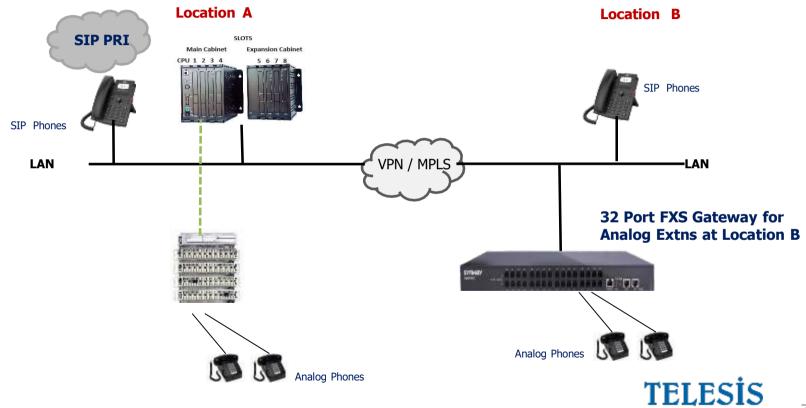
Connect IP Phone at Remote Location over Intra-office Network





DETAILS OF SOLUTIONS

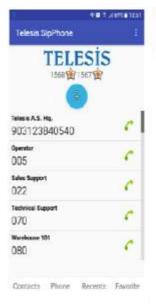
D Connect FXS Gateway at Remote Location over Intra-office Network



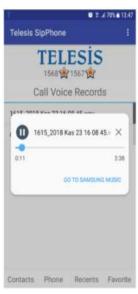


MOBILE SIP CLIENT











- · Supported by Android 6.1 or newer version
- · High voice quality
- · Dual SIP accounts
- Intelligent LAN and WAN ip address selection for data usage
- · Operation in both mobile and WiFi network
- · Manual mobile network selection even WiFi is available and connected
- · Bluetooth voice support and call control from BT device
- · Call waiting
- · Call transfer
- · New call through the first account when there is a waiting call
- · New call through the second account when there is a waiting call
- · Call voice recording
- · Access to recorded calls within the application
- · Manual sharing and mailing recorded call
- · Call deflection
- · Address book integration
- · Call detail records (CDR)
- · Prefix definition for PBX trunk calls
- · Trace of incoming, outgoing, missed and lost calls
- · List of favourite contacts





TELESIS is brand neutral and supports all end Point phones and extensions

- Analog Extensions
- IP Phones
- Soft phones Mobile App
- Desktop App
- Video Phones
- Portable Wi-Fi Phones.

Our Solution supports all

Analog and generic SIP

Phones of any make or

Model, including Wired,

Wireless and Portable Wi-Fi

Handsets



All Entry Level SIP Phones support Display, Caller ID, Transfer, Conference...

Sp-Phone, Call



Fanvil 301PV2, 10/100 switch



Fanvil 303G GIGA Bit switch Phone



SNOM D120 10/100 switch



SNOM D150, GIGA Bit switch Phone



Executive Level SIP Phones , incorporate Large Display, HD Quality Sp-phone and Memory Keys





SNOM D785





FANVIL W611W (Portable Wi-Fi Phone)





Operator Console Phone provides quick access to Memory keys and extension status

Highlights:

- Local phonebook, caller ID, call hold, call transfer
- HD audio on speakerphone and handset
- Can handle a variety of audio applications, including IP
 - voice, Multiparty conference, etc.
- Fast Ethernet 10/100 Mbps Ethernet with PoE enabled.
- Stand with 2 adjustable angles.
- BLF with Light indicator
- Navigation Keys
- Soft Function Keys



