

For More Info Call Atlas Telephones: (888) 689-1711







The G1E TransTel IP Communication System combines the advantages of traditional telecommunications with the convergence of VoIP technology. The system offers a vast array of flexible features to handle your business communication needs-Today and Tomorrow!

The base platform is designed for most small businesses, right out of the box. The G1E is initially equipped to accommodate up to eight digital system phones, all of which are thoughtfully designed to make the most of your productive day.

For customers who need mobility, the base system is also configured to accommodate up to four industry standard cordless telephone sets. There is even an embedded four channel automated attendant so your customers can dial station users or departments directly. You may even record your own custom on hold message for calls that are placed on hold in order to provide that professional image you are looking for!

Flexibly communicate on your own terms with a variety of outside line choices. The G1E offers a unique combination of outside line interfaces. It can serve as a branch office system for VoIP multi-site applications. Or the choice of efficient ISDN or PSTN (Public Switched Telephone Network) lines is available. Or a combination of the above can be used. Whatever your needs, the G1E offers a variety of ways to communicate:

- PSTN (Public Switch Telephone Network)
 Traditional and familiar loop start telephone lines. Caller ID is supported.
- ISDN (Integrated Service Digital Network) BRI (Basic Rate Interface)
 G1E has ISDN BRI with integrated Caller ID, Call Charge Metering Information, Direct Inward Dialing and MSN (Multi-Subscriber Numbers). Using ISDN a single outside line can have multiple incoming telephone numbers.
- VoIP (Voice-over Internet Protocol) SIP (Session Initiation Protocol)
 VoIP is the latest technology for voice communications. VoIP services can bypass the traditional telephone company and route voice traffic over an IP (Internet Protocol) based service, saving expensive toll costs. The G1E has an optional in-skin VoIP gateway that can be integrated to connect with IP Carriers for long distance voice traffic. This gateway also becomes the perfect solution for multi site business to business applications. In additional to toll bypass and cost savings, companies realize productivity gains via direct station to station dialing through a virtual private network.

Valuable Voice Service features enhance call handling

All G1E systems are equipped with a standard Auto Attendant and recordable Music-on-Hold feature.

Auto Attendant

Increases the standard 4 ports to 6 automated attendant ports to answer incoming calls simultaneously.

Voice Mail (Optional)

Every station has its dedicated voice mail box to let callers leave messages when you are busy or away from your desk. A Virtual Mail Box capability allows you to assign mailboxes for associates who do not have a physical phone in the system, or to record and play special product promotion announcements.

Conference Groups

Every conference group has maximum 8 participants. There are up to 5 groups when each group has 3 persons.

Conference Recording

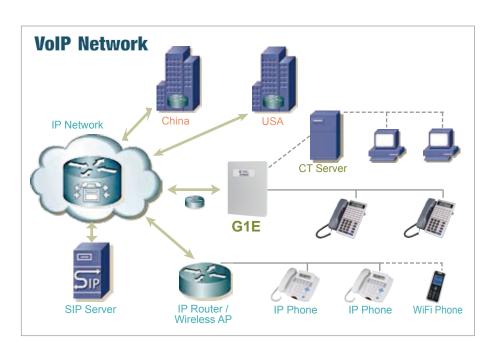
Record conference calls when needed.

Answering Machine Emulation

Listen to caller leaving a message in your mailbox and answer the call by just lifting the handset.

Live Call Recording

Record important or malicious conversations when needed. On demand as required or every call automatically.





CTI solution Increases Office Productivity

CTI (Computer Telephony Integration) applications employ your customer database to deliver caller information to your staff along with the call. G1E third party TAPI server, TSP (Telephony Service Provider), Caller ID and CTI applications integrate to deliver your business contact's information immediately to your desktop when they call you. CTI can be configured in many ways to suit your work, from a simple balloon to a full screen pop for your incoming call.

- Account Code Capability
 Access Control Phone*
 Answering Machine Emulation*
 Automatic Answer Intercom
 Auto Attendant with Overflow
 Auto Redial
 Busy Lamp Field
 Barge-in

- Barge-in
 Calendar and Real Time Clock
- Call Accounting (In Skin)

- Camp On
 CLI on Both External and Internal Calls
 Conference*
 Conference Recording*
 CTI Integration* TAPI
 Day and Night Service
 Group Listening

- Live Call Recording* Last Number Redial

- Last Number Redial
 Macro Keys
 Music On Hold Recordable
 On-Call Programming
 Paging/Internal and Meet Me
 Paging/External*
 Pick-up
 Privacy Release

- Remote Software Upgrade by Modem*

- Station Group
 SLT Station Caller ID Service
 SLT Industry Standard 90V Message Light Compatible
 SMDR (Station Message Detail Record)
 SMDR Real Time Protocol for Call Limitation
 Speed Dial System (900 Sets Max.)
 Speed Dial Station (20 Sets Max.)

- Voice Mail*
 VolP Call Point-to-point*
 VolP Call Point-to-multi-point*
 VolP Call DID Calls*
- *: Optional equipment and cost will be required for this feature.

G1E System Specifications & Capacities	
	Maximum ports
Analog CO Line	8
ISDN BRI	6 (12 channels)
VoIP Trunk - SIP	6
Digital Key Station	24
Single Line Telephone	20
Access Control Phone	24
Auto Attendant - Basic/Optional	4 / 6
Voice Mail - optional channel	4
KSU Dimension - LxWxH (mm)	425x315x90 mm



Trans*Tel* Partner:

TransTel Communications, Inc.

1562 Park Lane South, Suite 300 Jupiter, Florida USA 33458 U.S.A. Toll free 800-872-6785 Phone: 561-747-4466

Fax: 561-747-4578

PT. TRANSINDO INFOTEK

Kamal Raya Outer Ring Road Taman Palem Lestari Blk D 10 No. 3 Cengkareng-Jakarta Barat 11730 Tel: 62-21-5555788

Fax: 62-21-5555789

Design and specifications are subject to change without notice.