

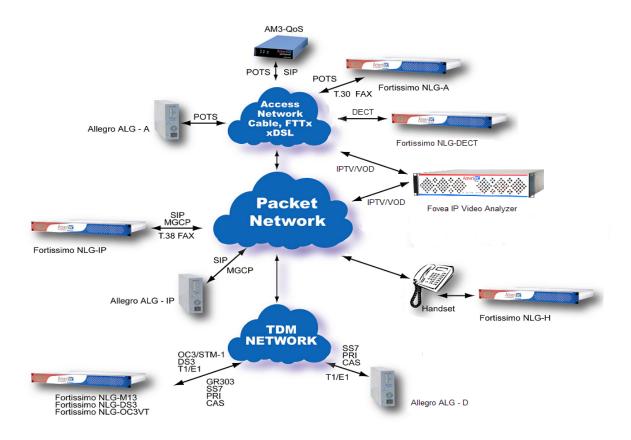
Allegro Desktop Network Call Generators



- ♦ Allegro is the world's smallest, full-featured network load generation product designed to test and qualify the voice and packet network.
- ◆ Allegro speeds up your system integration, functional and performance testing at an unbeatable price/performance ratio.
- ♦ Allegro is designed to place calls to or receive calls from other Allegro, Fortissimo or AM3-QoS units to provide a complete test system of call generation and Quality of Service (QoS) measurements.

The Allegro family of network load generators represents a significant technological advancement in desktop call generation. Designed to meet the requirements of today's developer of both legacy and new generation switches including hardware and software modules. Operators focusing on system integration or revenue and quality assurance will also enjoy the functionality, ergonomics and costs of this solution. The Allegro packs the power of 8 simultaneous SIP/MGCP calls, up to $4 \times T1/E1$ spans or $4 \times analog$ POTS (loop/ground/fax/data modem) circuits in a unit that sits inconspicuously on a desktop.

The Allegro comes in three versions - an Ethernet VoIP (SIP/MGCP) configuration, a multi-span T1/E1 configuration, or a 4-line analog configuration. The Allegro IP Voice Quality Tester (VQT) is capable of generating signaling and media traffic on 8 unique IP sessions. The QoS measurement package is standard on this unit. The digital version ships complete with 2 T1/E1 software selectable spans, a full quality of service (QoS) measurement package and CAS, PRI and SS7 signaling capabilities. Optionally, the unit may be configured for 3 or 4 digital spans. The analog version supports 4 analog circuits, complete with loop and ground start signaling, fax and data modem operation and Ameritec's industry leading QoS measurement package.

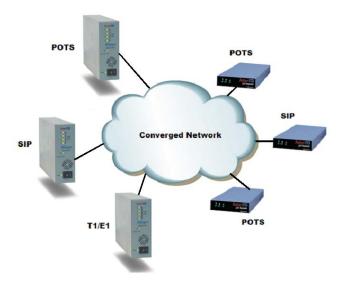




Applications

- Load Generation
- ◆ Cable Telephony, FTTP, PON, VoDSL
- ♦ Automated Testing via Telnet and Tcl
- Fax and Data Modem Testing
- Quality of Service Testing
- CLASS Calling Feature Testing
- Live Network Testing
- Point of Sale Terminal Testing
- Voicemail, IVR Testing
- Echo Testing
- Wideband Voice Testing and Verification

Live Network Testing



- Provides Call Detail Records with Timestamps
- Calling #, Called #, QoS Measurements
- Active Call Completion and QoS Testing

Operation

The Allegro products allow the developer to simulate various types of network traffic on a select number of spans or lines without having to go to the expense of purchasing large-scale units. Maintaining full testing flexibility on a smaller line count is the premise behind Allegro. Each Allegro is controlled via a PC workstation over a 10/100 MB Ethernet port and utilizes a Conductor graphical user interface (GUI) that is intuitive and easy to use. Alternately, a command set option allows the user to remotely control units via a Telnet or Tcl session for test automation applications. Full scripting of call scenarios is accommodated and manipulation of signaling protocols is provided to allow the user to fully test and debug equipment under test before releasing product to the next stage of development. Audio monitoring and capture is provided on the Allegro to complete the testing phase so that a developer can further verify the integrity of any call scenario.

Measurements

Long known for providing the most comprehensive set of measurements in the call generation industry, Ameritec has incorporated into every Allegro product the ability to simultaneously measure traditional call statistics such as calls originated and completed as well as a comprehensive set of QoS measurements designed to provide packet quality and voice quality scoring. This complete set of measurements provides the developer with PSTN to packet network correlation scores. The ability to utilize our GoldenVoice technology to measure packet loss, jitter, signal to noise ratio, clipping and delay while simultaneously providing R-factor, GMOS, G-PSQM, G-PESQ, and G-PESQ-LQ scoring on all lines makes this product an invaluable development tool.

Enhanced Features

A modem QoS feature verifies accurate detection and transport of Fax, Modem and TTY/TDD signals in pass-through scenarios. Synchronized call scripts provide the ability to test complex call scenarios such as A->B;B->A, and CLASS features such as conference calling, call waiting, etc.

Full wideband audio testing allows the Allegro IP to ensure that High Definition (HD) voice quality is operational.

Real fax and data modem emulation is fully supported on Allegro analog units to provide complete network testing.

Configuration

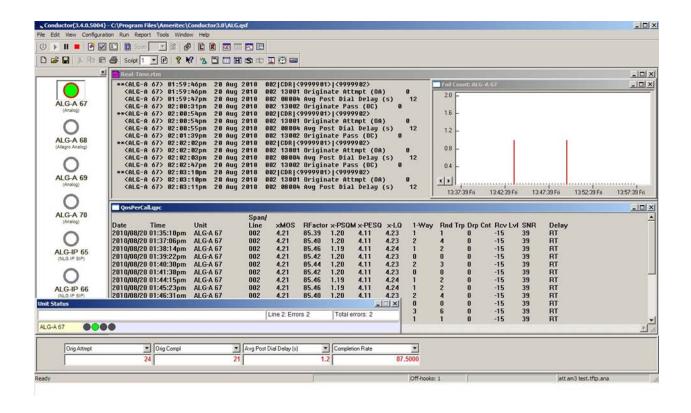
Upon power up, each Allegro prompts the user to recall previously stored test configurations from the PC. This includes the type of call programs that the user desires to run and any parameter settings that are required for a specific test.

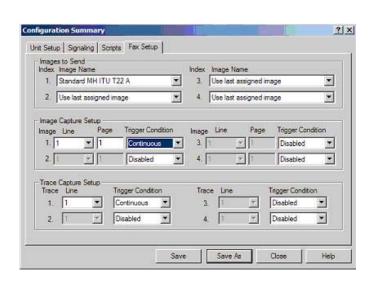
At the completion of any test, the user is prompted to store the unit's configuration so that it may later be retrieved to verify test integrity or re-run a previous test. Virtually, an unlimited number of these configurations can be stored on the workstation for easy recall.

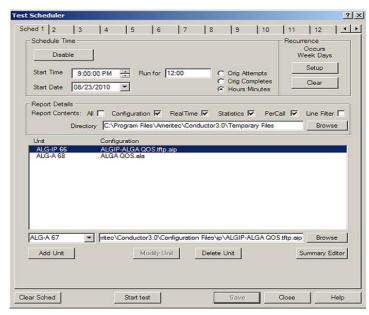
Performance

While this may be the industry's smallest network load generator, ounce for ounce it provides highest performance. Utilizing independent resources, each Allegro can generate as many calls as your equipment can support. The Allegro can configure nearly every parameter associated with the generation or answering of a call and therefore it is possible to test for load related issues as well as fundamental integrity issues. Additionally, as many as 32 Allegro units can be used side by side to increase call volumes without significantly increasing your investment.

Conductor Graphical User Interface







General Characteristics

User Interface Windows 7, Vista, XP based GUI

Ethernet Port One RJ45 connector and two LED

indicators per network interface

10/100 BaseT interface Complies with IEEE 802.3

Dimensions 2" Wide x 5" High x 7" Deep Power 90 – 264 VAC, 47 to 65 Hz

Weight 2 Pounds (0.9kg)

Call Programs and Scripts

Commonly used scripts supplied Features

with unit

Scripts created and downloaded

from workstation or PC

System

Capacity 8 unique IP sessions, 4 analog

lines, or 2 digital spans (opt. 4)

Call Volume Typically 500 confirmed calls per

> hour per line (DTMF dialing, tone ID confirmation, 2 unpaired lines)

Line Types - SIP/MGCP calls

> - Loop start/ground start, 2 wire - T1/E1 spans (CAS, PRI, SS7) - Pulse, DTMF, MFR1 & MFR2

dialing

Test Interface RJ45 (VoIP VQT), RJ45 (digital)

RJ11 (analog)

Fax and Data Modem

Data Modem Rates from 300 bps to 33.6 Kbps

Bell 103, 212, V.21-V.34

modulations

Connection rate, data throughput, Retransmissions, errors, error rate,

Signal level, carrier loss

Fax T.30 fax supported to V.17 rates

(14.4 K)

ECM, non-ECM Capture and trace **Voice Channel Functions**

Tone detectors are based on digital Detectors

signal processors (DSP) 1 per

channel

Call Progress Detectors One detector per line

> Detects dial tone, busy, reorder, ring, ringback, supervision

Path Confirmation One receiver per line

Accuracy: 1%, +/- 10 Hz

Single Frequency Tone

Generator

64 Selectable tones

Voice over Packet

Voice Path Confirmation GoldenVoice signal designed to pass

through vocoder

Packet Drop Out Count Count lost packets for frame sizes of 5,

10, 15, 20, 30, 40 and 100 ms

Round trip delay +/- 10 ms accuracy Measure Delay

One way delay +/- 5 ms accuracy

Measure Clipping Peak and average clipping of standard

reference with +/- 5 ms accuracy

Jitter Peak and average jitter of standard

reference with +/- 5 ms accuracy

SNR Average and maximum SNR received

(from 0 to 39 dB)

Signal Energy Average and maximum GoldenVoice

energy received (0 to -50 dBm)

Maximum non-GoldenVoice energy Spurious Energy

(0 to -50 dBm)

Total Energy Average and maximum GoldenVoice

energy plus extraneous noise received

(0 to -50 dBm)

GoldenVoice Noise received from 0 to -50 dBm

> Ten programmed GoldenVoice tone signals used to send the ID from each side encoded as three tone sequences

Calculation of R-Factor, GMOS, G-QoS

PSQM, G-PESQ

R-factor is based upon E-Model in ITU-T G.107, Amendment 1, June 2006

Printout and Reports – Call Statistics

Call Statistics for each line Manual Reports

Totals for all lines

Automatic Reports Prints automatically at 1 minute to 1

hour intervals

Contents of each column in the printout

are user selectable

Call Attempt Count Call Completion Count Delayed Start Signal Count No Start Signal Count No Alert Signal Count No Voice Path Count Busy Signal Count No Answer Count Average Dial Tone Delay Average Post Dial Delay Custom Code Report Count	GMOS tests	Average and maximum percent drop Drop packet size (ms) Drop test time (s) Average and maximum one-way delay (ms) Average and minimum circuit noise Minimum and maximum receive level Average and maximum round trip delay (ms) Average and maximum SNR
Attempted Calls Count Completed Calls Count Custom Code Report Count	Analog Impedances Supported (ALG-A+)	600 ohm, 900 ohm 600 ohm + 1uF, 600 ohm + 2.16uF, 900 ohm + 1uF, 900 ohm + 2.16uF, 270 ohm + (750 ohm 150nF), 220 ohm + (820 ohm 120nF), 370 ohm + (620 ohm 310nF),
VoP Statistics Average and maximum dropout duration (ms) Average and maximum front clip duration (ms) Average and maximum back clip		320 ohm + (1050 ohm 310hr), 320 ohm + (1050 ohm 230nF), 370 ohm + (820 ohm 110nF), 275 ohm + (780 ohm 150nF), 120 ohm + (820 ohm 110nF), 350 ohm + (1000 ohm 210nF), 0 ohm + (900 ohm 30nF)
duration (ms) Average and maximum jitter duration (ms)	Ordering Information Mainframe/Chassis/Test	Allegro 4 line Analog Call Generator
Average and maximum dropout duration (ms) No tone detected Tone lost	ALG-D	Allegro 4 line Analog Call Generator e/w ability to support Data Modem and/or Fax Allegro 2 span Digital T1/E1 Call Generator
Average, minimum and maximum one- way delay (ms)	Options	Allegro 8 IP session SIP/MGCP Call Generator
Minimum and maximum total energy Minimum and maximum SNR Minimum and maximum GoldenVoice energy Maximum spurious energy overflow Spurious energy overflow Low SNR	250532 250533 250534 250538 250566 250567	Digital License Key, adds 2 T1/E1 spans to ALG-D unit Command Set Feature G.729 Software Analog GPS Option Extended Feature Set Fax License for ALG-M (2 lines max) Modem License for ALG-M (2 lines
Average, minimum and maximum round trip delay (ms)	250571 Accessories 190005	max) Tcl License Protocol Development Kit XpresScript Visual Scripting Tool
	Call Completion Count Delayed Start Signal Count No Start Signal Count No Alert Signal Count No Voice Path Count Busy Signal Count No Answer Count Average Dial Tone Delay Average Post Dial Delay Custom Code Report Count Attempted Calls Count Completed Calls Count Custom Code Report Count VOP Statistics Average and maximum dropout duration (ms) Average and maximum back clip duration (ms) Average and maximum jitter duration (ms) Average and maximum dropout duration (ms) Average and maximum dropout duration (ms) Average and maximum jitter duration (ms) Mo tone detected Tone lost Average, minimum and maximum one- way delay (ms) Minimum and maximum GoldenVoice energy Maximum spurious energy overflow Spurious energy overflow Low SNR Average, minimum and maximum	Call Completion Count Delayed Start Signal Count No Start Signal Count No Voice Path Count Busy Signal Count No Answer Count Average Dial Tone Delay Average Post Dial Delay Custom Code Report Count Attempted Calls Count Completed Calls Count Custom Code Report Count VoP Statistics Average and maximum dropout duration (ms) Average and maximum back clip duration (ms) Average and maximum jitter duration (ms) Average and maximum dropout duration (ms) Average and maximum dropout duration (ms) Average and maximum jour duration (ms) Average and maximum dropout duration (ms) Average, minimum and maximum one- way delay (ms) Minimum and maximum SNR Minimum and maximum Golden Voice energy Minimum and maximum Golden Voice energy Maximum spurious energy overflow Low SNR Average, minimum and maximum round trip delay (ms) Accessories

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