

# MultiDSLAs nodes datasheet

**This Datasheet describes features, specifications and ordering information relating to the Node types which may be part of a MultiDSLAs test system. A complete test system consists of a MultiDSLAs Controller user interface application, plus one or more types of the 'node' devices described here**

See also the following:

MultiDSLAs Controller Datasheet, for details system features

MultiDSLAs Brochure, for a general description of the MultiDSLAs system

Audio Streaming Integrity Brochure, for details of this option

## Node Selection Guide

- ▶ DSLA Series / Analog – Use for testing involving cellular phones, desk phones, analog (POTS/PSTN) phone lines, ATA's, PC sound cards (for soft phones), audio streaming devices...
- ▶ VPP Series – Use for VoIP network testing, VoLTE handset evaluation and in any testing

## Node type—Quick reference

Model	Type	Interface	Hardware	Software	No. of Nodes
DSLAIIC	Analog	RJ-22, RJ-11, 4mm Balanced	Desktop	(Measurement and control firmware in device)	2
DSL3	Analog	RJ-22 and 3.5 mm jack handset module	Modular 19" rack mount	(Measurement and control firmware in module)	1 to 6 modules
		RJ-11 POTS module			
VPP-fn	VoIP/SIP	Ethernet	No	Windows service	1-5, 10, 20, 30, 50, 100
VPP+n-f					

## DSL A Series / Analog Nodes

DSL A Technical Specification:

Dimensions (mm): **DSLAIIC** 72h x 218w x 200d

Net weight: **DSLAIIC** approx 3kg

Power: **DSLAIIC** 100-240Vac (external PSU) or 9-18Vdc, 12W

Operating temperature range: -2 to +40°C

Approvals & Compliance: CE Mark; FCC47 CFR Part15

Calibration: full calibration report supplied;  
recommended re-calibration cycle 3 years



### Test Signal Generation

- DSL A Series / Analog – Use for testing involving cellular phones, desk phones, analog (POTS) phone lines, ATA's, PC sound cards (for soft phones), audio streaming devices...
- Any user-supplied speech material in wav or PCM format, generated with user-defined mean active speech level with setting range -99dBm to +10dBm
- Artificial Speech Test Stimulus (ASTS) British or American English; 8k and 16k sample rate
- Sine wave 20Hz to 22kHz, setting range -99dBm to +10dBm, any duration
- Swept sine wave 20Hz to 22kHz, setting range -99dBm to +10dBm, any duration
- White, Gaussian white or pink noise, setting range -99dBm to +10dBm, any duration
- DTMF setting range -99dBm to +10dBm any duration
- 6 Method B
- Noise in speech to within 20dB of mean active speech level
- Noise in speech to within 20dB of mean active speech level
- Peak and True RMS Levels
- Units of measurement dBm, mV
- Tone burst measurement mode
- Measurement of doubletalk (percentage of measurement period where speech is present on both channels)
- Linearity 0.1dB for levels -60 to +10dBm
- Linearity 0.1dB for frequencies 20Hz to 22 kHz □ Noise floor -85dBm or better
- Range of measured levels -75dBm to +19dBm
- Minimum measurable mean active speech level -65dBm
- Dynamic range of 4-wire inputs 110dB
- Synchronization
- GPS (product option) - GPS time and position data
- Network Time Protocol (NTP)

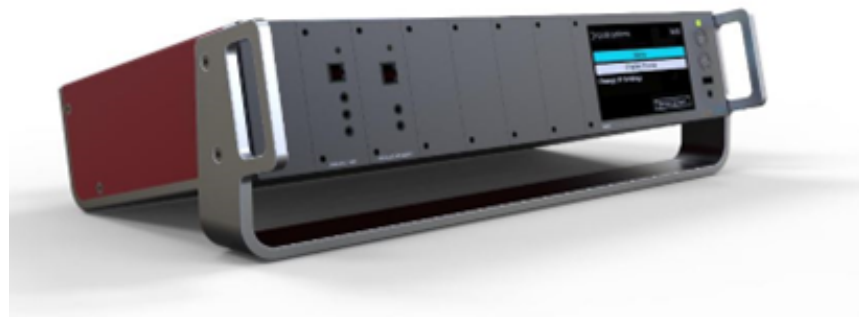
#### DSL A Series Accessories

- GPS module
- Bluetooth adapters for Narrowband and Wideband speech, and audio streaming
- Universal Smartphone adapters with LRGM and LRMG pinouts
- DSL A Connection Cables - two sets of cables to link
- DSL A to PC and laptop sound cards

## DSLA Series / DSLA 3

**DSLA3 is the new hardware for analog testing driven by MultiDSLA platform.**

**This device is a 6-slot modular system that allows users to flexibly define which interfaces to use. Build your own test system specifically according to your needs.**



The modular conception allows a wide choice of connections such as PSTN lines, phone handsets, legacy adapters used with DSLA2c for Bluetooth, PTT, and any kind of mobile devices. DSLA3 also accepts legacy GPS modules.

DSLA3 embeds a Wi-Fi 5 module to either connect to your network or create an access point to relay the Smartphone control command to the tested Smartphones.

DSLA3 includes a touchscreen to configure its IP and MultiDSLA IP/hostname to reach.

### Test Signal Generation

- Signal sampling rate capability up to 48kHz
- Any user-supplied speech material in WAV or PCM format, generated with user-defined mean active speech level with setting range -99dBm to +10dBm
- Sine wave, including swept and noise 20Hz to 22kHz, setting range -99dBm to +10dBm, any duration
- DTMF setting range -99dBm to +10dBm, any duration
- DTMF user-defined twist, frequency offset and break duration
- Conversational speech with/without double-talk
- Two independent tracks on each DSLA channel to create Complex mixed signals, e.g. speech plus noise

### Measurements

- Linearity 0.1dB for levels -60 to +10 dBm
- Linearity 0.1dB for frequencies 20Hz to 22 kHz
- Noise floor -85dBm or better
- Range of measured levels -75dBm to +19dBm
- Minimum measurable mean active speech level -65dBm

### Module Specifications

- ▶ Handset module
  - 4 wires RJ22 interface
  - Control and sync 3.5 mm female jack
  - Smartphone 3.5 mm female jack LRGM/LRMG
  - Floating inputs (10 k $\Omega$ ) and outputs (25  $\Omega$ )
  - Output level attenuated by 28dB
- ▶ Line module
  - 2 wires RJ11 FXO interface
  - Control and sync 3.5 mm female jack
  - Phone line ports 600  $\Omega$  or complex impedance
  - Output level limited to +6dB
  - DTMF or Pulse dialing, Caller ID on line without adapter

Dimensions: 89Hx244Wx440D (2U)

Net weight: 5Kg

Power: 100-220 V AC

Power Frequency 50-60 Hz

Power consumption: 2.3 A max.

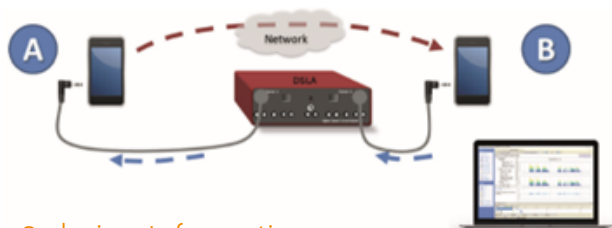
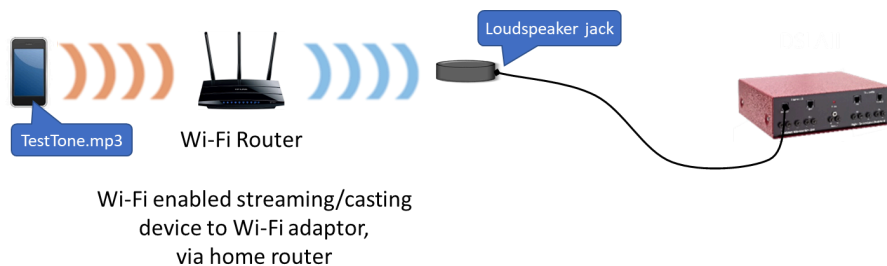
Operating Temperature: -2 to 40 °C

Approval and compliance: ongoing

Calibration: every 3 years for analog modules

## DSLAI Use cases

### Audio Streaming Integrity



### Cellular Voice Quality Testing using DSLAI

### Ordering Information

Product No.	Model	Description
<b>Digital Speech Level Analyser</b>		
<b>DSLAI</b>		
000029	DSLAIIC	DSLAIIC - 2 channel unit
<b>DSLAI3</b>		
000165	DSLAI3	DSLAI3 - Chassis
000166	DSLAI3-MO-AL	RJ11 PSTN FXO Analog Line Module
000167	DSLAI3-MO-AH	RJ22 Analog Handset Module
<b>DSLAI Options and Accessories</b>		
<b>BlueTooth</b>		
000011	APT	Custom Bluetooth adaptor, NB/WB speech, APTx Low-Latency
000012	USP	Universal Smartphone LRG Adapter
000013	USPR	Universal Smartphone LRM Adapter
000224	USBC	USB Type C to Jack 3.5mm adapter for LRG/LRM
<b>GPS</b>		
000016	GPSM-USB	GPS Module - USB power supply connector
000017	GPSM-DSLAI	GPS Module - DSLAI power supply connector
000018	GPSM-SERIAL	GPS Module - DSLAI Serial Connector (from DSLAI S/N 5945)
000019	GPS-E25	GPS Extension Cable - 12V version - 25m for GPSM-DSLAI
000020	GPS-E25S	GPS Extension Cable - 5V version - 25m for GPSM- SERIAL
000021	GPS-E50	GPS Extension Cable - 12V version - 50m for GPSM-DSLAI
000022	GPS-E50S	GPS Extension Cable - 5V version - 50m for GPSM- SERIAL
000014	GPSCONV-USB	GPS Connection Cable Conversion for supplied Garmin GPS - USB power supply
000015	GPSCONV-DSLAI	GPS Connection Cable Conversion for supplied Garmin GPS - DSLAI power supply
<b>Other Accessories</b>		
000145	DSLAI POWER SUPPLY	Power Supply for DSLAIIC (part number 000029)
000023	DCC	DSLAI Connection Cables
000139	CID	Caller ID Cable Accessory
<b>DSLAI Upgrades</b>		
000024	DSLAI48kUPG	DSLAIIC upgrade for 48k sample rate support

## VPP Series / VoIP Nodes

Vox Port Packet is the reference softphone within MultiDSLA systems. VPP is used in labs for mobile tests with base stations and as a SIP service testing tool, allowing to monitor service availability and performance, on premise or for cloud-based solutions.

Licensing is managed by MultiDSLA controller. 'f' in VPPf or VPP+f means floating license.

Item	VPPf	VPP+f
Requisites	Windows 10 / 11	
	Windows Server 2016 / 2019	
	Intel Core Duo, 2 GB RAM minimum	
Network Interface (NIC) and IP Management	Definable network test interface for each call	
	IPv4 / IPv6 support	
Codec Support	G.711, G.729, G.729A, G.729B, G.723.1, G.722, G.726, iLBC, Opus, 8k, 16k, 32k linear pcm	
More codec support	-	EVS all modes, AMR NB & WB with DTX,
Frame size	5, 10, 20, 30, 40, 50, 60ms codec dependent	
Parallel instances per VPPf host	Maximum 30,	
	Can go up to 50 with specific hardware requisites	
User-defined static jitter buffer	X	X
Signaling capture	X	X
SIP over UDP	X	X
SIP over TCP		X
SIP over TLS		X
Secured RTP		X
IMS support		X
Inband and outband DTMF	X	X
SIPLess (no signaling, just RTP) support		X
DSCP tagging	X	X
Jitter and packet loss generation on output stream		X
AMR EVS in-call bit rate change		X
EVS to AMR interoperability		X
Packet based Loopback	X	X

Compliances

Signalling:

RFC2617

RFC2976

RFC3261

RFC3264

RFC3325

RFC3903

RFC4568

RFC5630

Media:

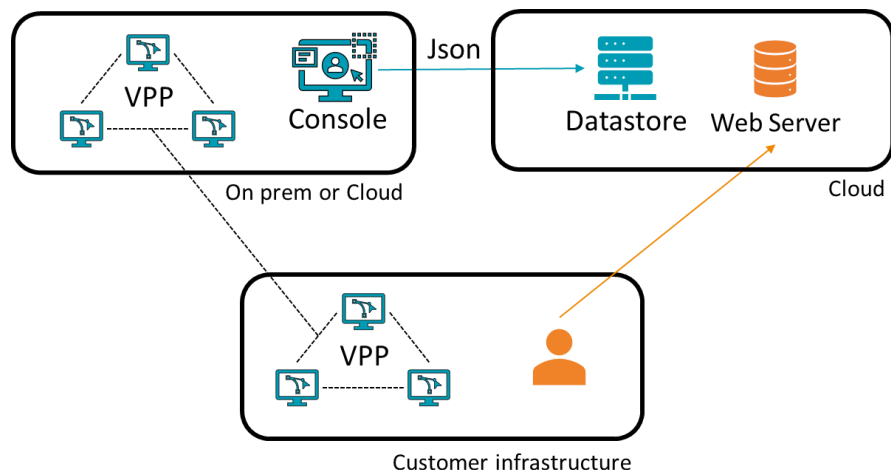
RFC3550

RFC3711

DTMF:

RFC4733

## MultiDSL A Vox Port Packet use cases



24/7 SIP service monitoring  
SIP quality and Root cause analysis  
100% software  
Cloud-based or on premise observability offer  
Speech to text capabilities for IVR tests



Drive testing  
Call are placed again if closed  
GPS location  
PESQ and POLQA scoring  
Signal Strength (SMC)

## Ordering Information

Product N°	Model	Description
Vox Port Packet Nodes one time licenses		
VPPf: Automatically re-assignable instances of VPP		
000210	VPPf-LIC	MultiDSL A Licence for VPPf Support - Includes 1x VPPf1 and 1x MUI-DS-1 node license
000184	VPPf-ADD-1	1 additional VPPf instance - Include 1x node licence for MultiDSL A (MUI-DS-1) - Requires VPPf
VPP+f: Automatically re-assignable instances of VPP		
000211	VPP+f-LIC	MultiDSL A Licence for VPP+f Support - Includes 1x VPP+f1 and 1x MUI-DS-1 node license
000189	VPP+f1-ADD-1	1 additional VPPf+ instance - Include 1x node licence for MultiDSL A (MUI-DS-1) - Requires VPP+f
Vox Port Packet Yearly subscription licenses		
VPPf		
000212	VPP-Setup	One-Time Setup Fee for VPPf
000213	VPPf1-Sub	1x VPPf yearly licence subscription for MultiDSL A - Subscription Mode - Yearly price
VPP+f		
000214	VPP+-Setup	One-Time Setup Fee for VPP+f
000215	VPP+f1-Sub	1x VPP+f yearly licence subscription for MultiDSL A - Subscription Mode - Yearly price