

# MultiDSL A nodes datasheet

This datasheet outlines the features, specifications, and ordering information for the various node types that can be integrated into a MultiDSL A test system. A complete system includes the MultiDSL A Controller user interface application along with one or more of the node devices described in this document.

See also the following:

MultiDSL A Controller Datasheet, for details system features

MultiDSL A Brochure, for a general description of the MultiDSL A system

## Node Selection Guide

- ▶ DSL A Series / Analog
  - Use for testing involving cellular 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
DSL A3	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					

# DSLAI Series / Analog Nodes

DSLAI Technical Specification: Dimensions (mm):

DSLAI 72h x 218w x 200d Net weight: DSLAI

approx 3kg Power: DSLAI 100-240Vac

(external PSU) or 9-18Vdc, 12W Operating

temperature range: -2 to +40°C Approvals &

Compliance: CE Mark; FCC47 CFR Part15

Calibration: report recommended re-calibration cycle 3 years



## Test Signal Generation

- DSLAI 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 P.56 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)

### DSLAI Series Accessories

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

## DSLA Series / DSLA 3

DSLA3 provides more density 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.



streaming  
Universal  
adapters wi  
DSLA Connection Cables - Smartphone  
two sets  
DSLA to PC and laptop sound  
cards GPS module  
Bluetooth adapters for  
Narrowband

DSLA3's modular design offers a wide range of connection options, including PSTN lines, phone handsets, and legacy adapters used with the DSLA2c for Bluetooth, Push-to-Talk (PTT), and various mobile devices.

The DSLA3 features an integrated touchscreen for easy configuration of its IP settings and graphical view of its configuration.

### 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
- 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

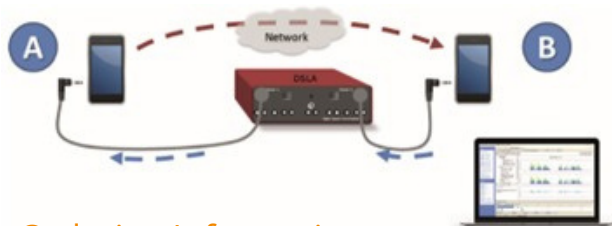
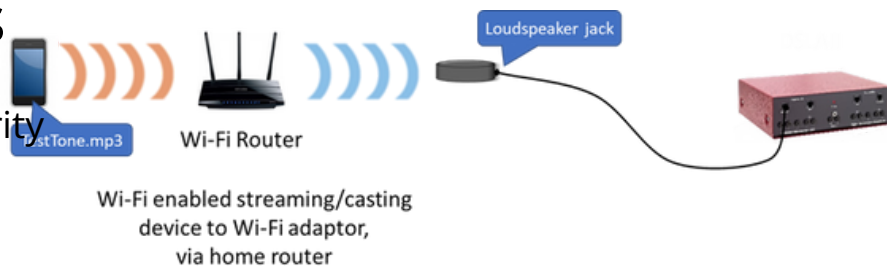
### Module Specifications

- ▶ 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



### Ordering Information

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

## VPP Series / VoIP Nodes

Vox Port Packet is the reference softphone within MultiDSLAs 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 MultiDSLAs controller. .

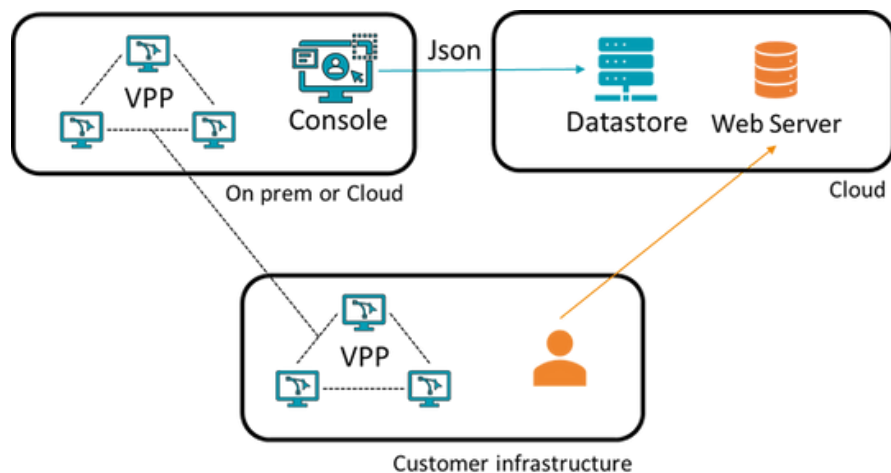
Item	VPPf	VPP+f
Requisites	Windows 10 / 11	
	Windows Server 2016 / 2019	
	Intel Core Duo, 2 GB RAM minimum	
Network Interface (NIC) and IP Management	Definable network test interface for each call	
	IPv4 / IPv6 support	
Codec Support	G.711, G.729, G.729A, G.729B, G.723.1, G.722, G.726, iLBC, Opus, 8k,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

Compliance  
Signalling:  
RFC2617  
RFC2976  
RFC3261  
RFC3264  
RFC3325  
RFC3903  
RFC4568  
RFC5630

Media:  
RFC3550  
RFC3711

DTMF:  
RFC4733

# MultiDSLAVox Port Packet use cases



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



Drive testing  
Call are placedagain 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	MultiDSLAVox Port Packet 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 MultiDSLAVox Port Packet (MUI-DS-1) - Requires VPPf
VPP+f: Automatically re-assignable instances of VPP		
000211	VPP+f-LIC	MultiDSLAVox Port Packet Licence for VPP+f Support - Includes 1x VPP+f1 and 1x MUI-DS-1 node license
000189	VPP+f1-ADD-1	1 additional VPP+f instance - Include 1x node licence for MultiDSLAVox Port Packet (MUI-DS-1) - Requires VPP+f
Vox Port Packet Yearly subscription licenses		
VPPf 000212 000213 VPP+f 000214 000215		
	VPP-Setup	One-Time Setup Fee for VPPf
	VPPf1-Sub	1x VPPf yearly licence subscription for MultiDSLAVox Port Packet - Subscription Mode - Yearly price
	VPP+-Setup	One-Time Setup Fee for VPP+f
	VPP+f1-Sub	1x VPP+f yearly licence subscription for MultiDSLAVox Port Packet - Subscription Mode - Yearly price