

# 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

- ▶ DSLAs 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
DSLAIIC4		RJ-22, RJ-11	19" rack-mount		4
DSLAIIC6					6
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

**DSLAIIC4/C6** 85h x 425w x 387d

Net weight: **DSLAIIC** approx 3kg **DSLAIIC4/C6** approx 7kg

Power: **DSLAIIC** 100-240Vac (external PSU) or 9-18Vdc, 12W **DSLAIIC4/C6** 100-240Vac

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

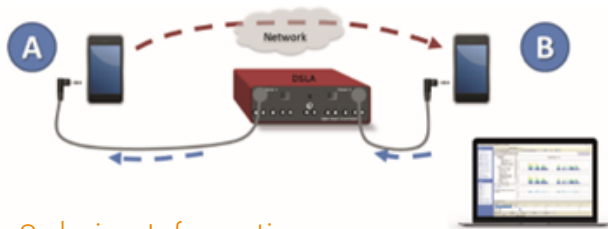
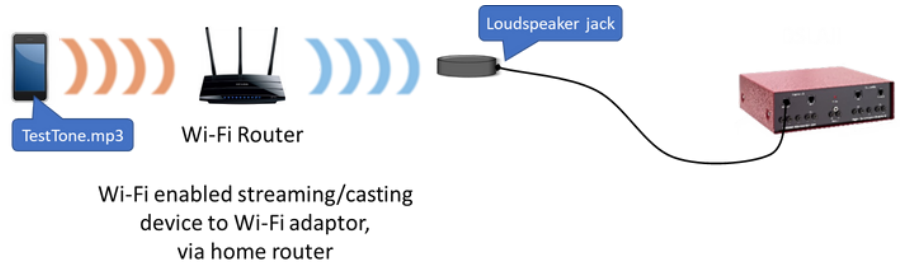
- DSL A Series / Analog – Use for testing involving cellular phones, desk phones, analog (POTS) phone lines, ATA's, PC sound cards (for soft phones), audio streaming devices...
- Any user-supplied speech material in wav or PCM format, generated with user-defined mean active speech level with setting 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 LRG M and LRM G pinouts
- DSL A Connection Cables - two sets of cables to link
- DSL A to PC and laptop sound cards

## DSL A Use cases

### Audio Streaming Integrity



### Cellular Voice Quality Testing using DSLA

### Ordering Information

Product No.	Model	Description
<b>Digital Speech Level Analyser</b>		
<b>DSLAI</b>		
000029	DSLAIIC	DSLAIIC - 2 channel unit
000030	DSLAIIC4	DSLAIIC - 4 channel 19 inch rack-mounting unit
000031	DSLAIIC6	DSLAIIC - 6 channel 19 inch rack-mounting unit
<b>DSL A3</b>		
000165	DSL A3	DSL A3 - Chassis
000166	DSL A3-MO-AL	RJ11 PSTN FXO Analog Line Module
000167	DSL A3-MO-AH	RJ22 Analog Handset Module
<b>DSL A Options and Accessories</b>		
<b>BlueTooth</b>		
000011	APT X	Custom Bluetooth adaptor, NB/WB speech, APTx Low-Latency
Handset		
000012	USP	Universal Smartphone LRG M Adapter
000013	USPR	Universal Smartphone LRM G Adapter
<b>GPS</b>		
000016	GPSM-USB	GPS Module - USB power supply connector
000017	GPSM-DSL A	GPS Module - DSL A power supply connector
000018	GPSM-SERIAL	GPS Module - DSL A Serial Connector (from DSL A S/N 5945)
000019	GPS-E25	GPS Extension Cable - 12V version - 25m for GPSM-DSL A
000020	GPS-E25S	GPS Extension Cable - 5V version - 25m for GPSM- SERIAL
000021	GPS-E50	GPS Extension Cable - 12V version - 50m for GPSM-DSL A
000022	GPS-E50S	GPS Extension Cable - 5V version - 50m for GPSM- SERIAL
000014	GPS CONV-USB	GPS Connection Cable Conversion for supplied Garmin GPS - USB power supply
000015	GPS CONV-DSL A	GPS Connection Cable Conversion for supplied Garmin GPS - DSL A power supply
<b>Other Accessories</b>		
000145	DSL A POWER SUPPLY	Power Supply
000023	DCC	DSL A Connection Cables
000139	CID	Called ID Cable Accessory
<b>DSL A Upgrades</b>		
000024	DSL A48kUPG	DSL AIIC 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. ‘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, EVS all modes, AMR NB & WB with DTX, 8k, 16k, 32k linear pcm	
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:

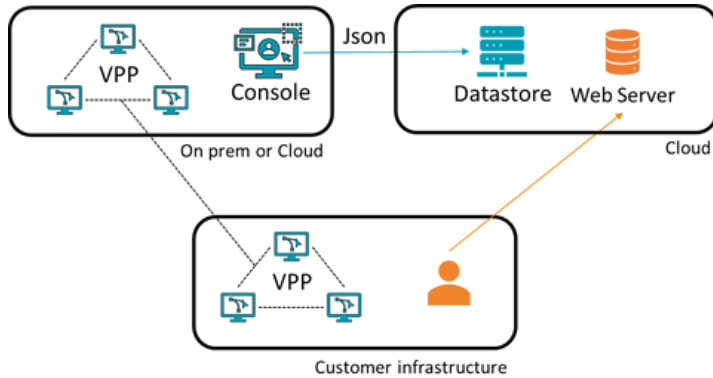
RFC3550

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 observability offer  
 Speech to text capabilities for IVR tests



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

## Ordering Information

Product N°	Model	Description
VoxPort Packet Nodes		
VPPf: Automatically re-assignable instances of VPP		
000184	VPPf1	1 floating re-assignable instance of VPP
000185	VPPf2	2 floating re-assignable instances of VPP
000186	VPPf3	3 floating re-assignable instances of VPP
000187	VPPf4	4 floating re-assignable instances of VPP
000188	VPPf5	5 floating re-assignable instances of VPP
000063	VPPf10	10 floating re-assignable instances of VPP
000064	VPPf20	20 floating re-assignable instances of VPP
000065	VPPf30	30 floating re-assignable instances of VPP
000066	VPPf50	50 floating re-assignable instances of VPP
000067	VPPf100	100 floating re-assignable instances of VPP
VPP+f: Automatically re-assignable instances of VPP		
000189	VPP+f1	1 floating re-assignable instances of VPP+
000190	VPP+f2	2 floating re-assignable instances of VPP+
000191	VPP+f3	3 floating re-assignable instances of VPP+
000192	VPP+f4	4 floating re-assignable instances of VPP+
000193	VPP+f5	5 floating re-assignable instances of VPP+
000076	VPP+f10	10 floating re-assignable instances of VPP+
000077	VPP+f20	20 floating re-assignable instances of VPP+
000078	VPP+f30	30 floating re-assignable instances of VPP+
000079	VPP+f50	50 floating re-assignable instances of VPP+
000080	VPP+f100	100 floating re-assignable instances of VPP+