



VDS-II

Intelligent VoIP Recording for standardized and proprietary protocols

With VDS-II VoiceCollect® developed a powerful VoIP Middleware which covers Standard Protocols and is able to integrate smooth and seamless into proprietary VoIP Communication Systems. VDS-II receives and processes VoIP communication and forwards content and call related data & events to the recording system.

- VoIP Decoding
- SIP, RTP & RTSP Protocols
- Alcatel, Thales, Siemens, Unify & Motorola connectivity
- Redundant solutions
- Highly scalable
- COTS Hardware with Microsoft® Windows® 7, Microsoft® Server 2008 R2 or 2012 operating system
- Deployment on virtual systems (VM Ware, Hyper-V, others)

VDS-II SIP

Recording according to SIP standard, available for passive or active connection.

Supports registration at SIP Registrar or SIP Proxy (SIP Active).

Supports SIP over UDP and SIP over TCP. Static Routing Function to allocate communication sources to recording channels.

VDS-II RTP

Recording according to RTP standard, available for passive or active connection.

Optional with VOX control. Supports Multicast registration (RTP Active).

Supported Codecs: G.711, G.723, G.726, G.728, G.729, GSM, iLBC and TETRA.

VDS-II Alcatel IP DR-Link

Active recording from Alcatel IP phones via DRLINK (Dedicated Recording Link) connection. Comfortable selection of extensions to record via marking tool.

Certified in line with the membership of VoiceCollect® in the Alcatel-Lucent Application Partner Program.

VDS-II ED137

Recording according to ED137 B part 4 standard for the use of VoIP communication in Civil Aviation.

Active connection to the recording system by other ED 137-enabled components like radio transmitters or voice communication systems (VCS) for single or redundant voice recording.

Static Routing Function to allocate communication sources to recording channels.

Storage and visualisation of events like PTT, Squelch or STD.

