



Migration to IP for Alarms

Markus Martinides, Overview December 2016





Existing Alarms



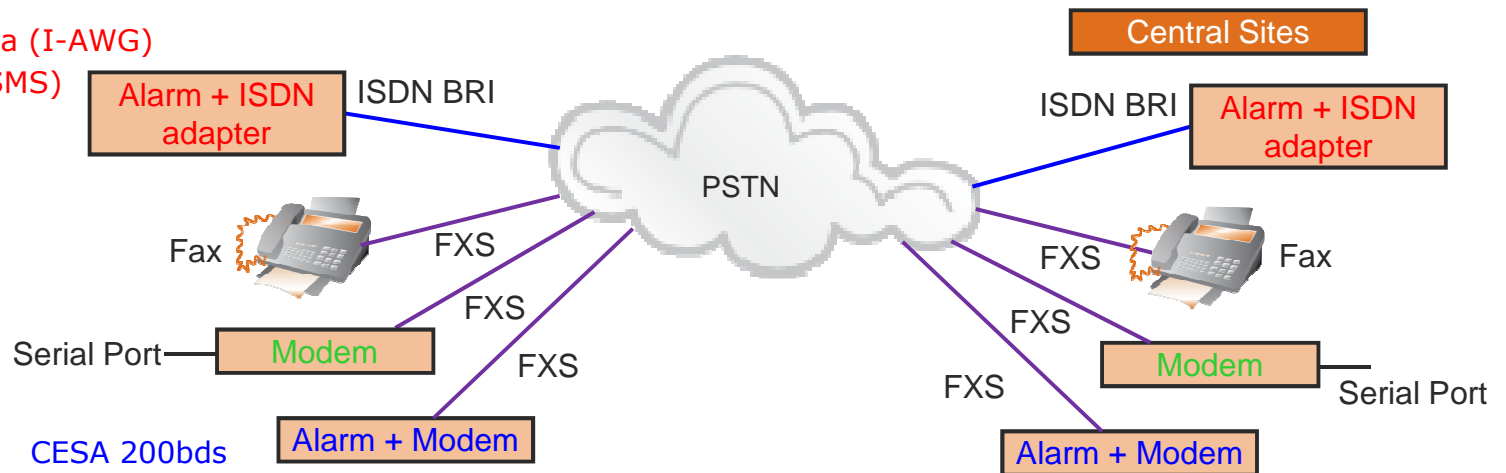
- All protocols are transported transparently over a switched communication
 - ISDN-Basic Rate Interface, or
 - FXS analog phone interface

- Alarmnet V.110 / Data (I-AWG)
- V.110 / UCP (Pager/SMS)

Remote Sites

- ESPA 444
- PSP (RS232)

- CESA 200bds
- Ademco Contact ID
- Ademco SIA
- ContactID DTMF
- FSK V21
- alarmnet 300 Baud / V.21 (AWG P3/P4)
- 4800 Baud / V.42bis (Pager/SMS)
- 14400 Baud / V.32bis (Fernwartung)

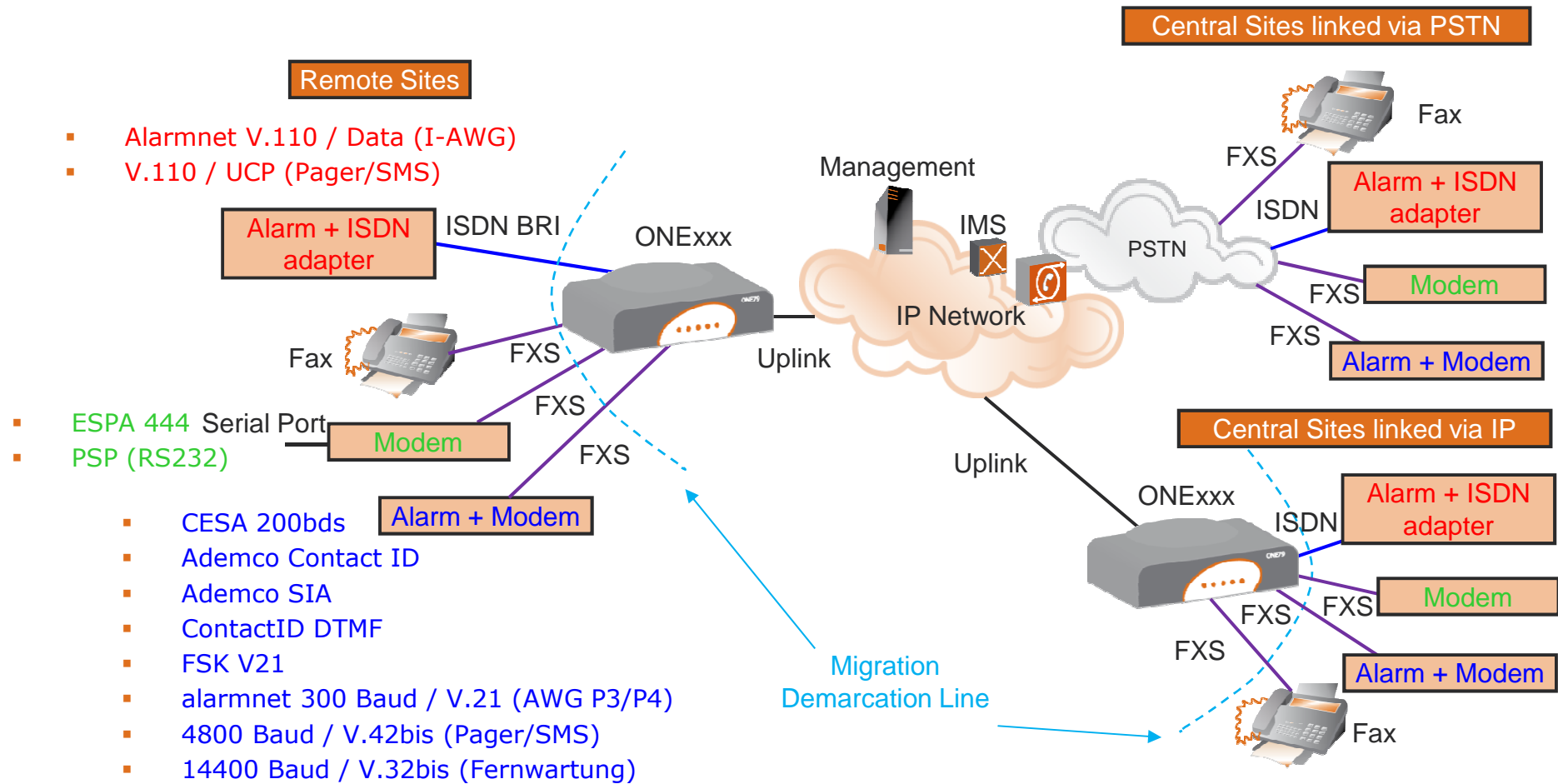




Migration from Existing Alarms



- All protocols are transported transparently over a IP network via ONExxx with
 - ISDN interfaces, and/or
 - FXS analog phone interfaces



- Alarmnet V.110 / Data (I-AWG)
- V.110 / UCP (Pager/SMS)
- ESPA 444 Serial Port
- PSP (RS232)
- CESA 200bds
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Technical Qualification



- What are the WAN technologies needed ? At which speed ?
 - SHDSL/bis, ADSL/2/+/m, VDSL2, Giga/FastEthernet Point-to-Point Fiber, Giga/FastEthernet UTP
 - 3G, 4G,....
- Are multiple WAN technologies needed ?
 - For backup, for load sharing,...
- How many ports/type per remote site and central site ? Point-to-point only ?
 - Which mix could help to optimize features vs cost ?
- Can the applications support a one-way delay 20-200 ms ?
- What about QoS in new IP networks used ?
 - Can we have Expedited-Forwarding QoS for application flows ?
- What OA clocking solutions would be the most adequate ?
 - 5ppm quartz, DSL-based, SIP RTP synchro,...
- Could all the needed protocols be managed with interfaces below ?
 - ISDN BRI NT
 - FXS dial-up
- Would these interfaces be needed ?
 - Analog Leased Line 2/4wires (300-3400 Hz / 600 Ω / RJ45)
 - I/O (digital 2-pole dry contact RJ45)
 - 1 async RS232 or ? Which connector ?
 - 1 sync RS232 or ? Which connector ?
- Which IMS/Softswitch ?
- Which management, provisioning ?
 - SNMP, TR-069



Comparison ONE vs ATA



	ONE425-4B-4V AV2 F4TEWn/a	Cisco ATA 190
FXS analog ports for modem, phone or fax	4	2
ISDN-BRI ports (ex: for V.110 alarms)	4	0
LAN ports	Four-port GigaEthernet UTP switch	0
Uplinks	One ADSL2+/VDSL2 port and One FastEthernet UTP/SFP port	One FastEthernet UTP port
64kbit/s UDI transparent transport	Supported	Not supported
T.38 fax relay and Passthrough G.711	Supported	Supported
Enhanced clocking managements (SIP RTP,...)	Supported	Not supported
<5ppm clock generation (for long fax)	Supported	Not supported
Echo cancellation	32ms	8ms ?
Management protocol CWMP	Supported	Not supported

- NB1: Cisco ATA 18x is EOL and replaced with ATA 190
- NB2: In addition to OA model below, other OA models exist with different uplink/user interfaces

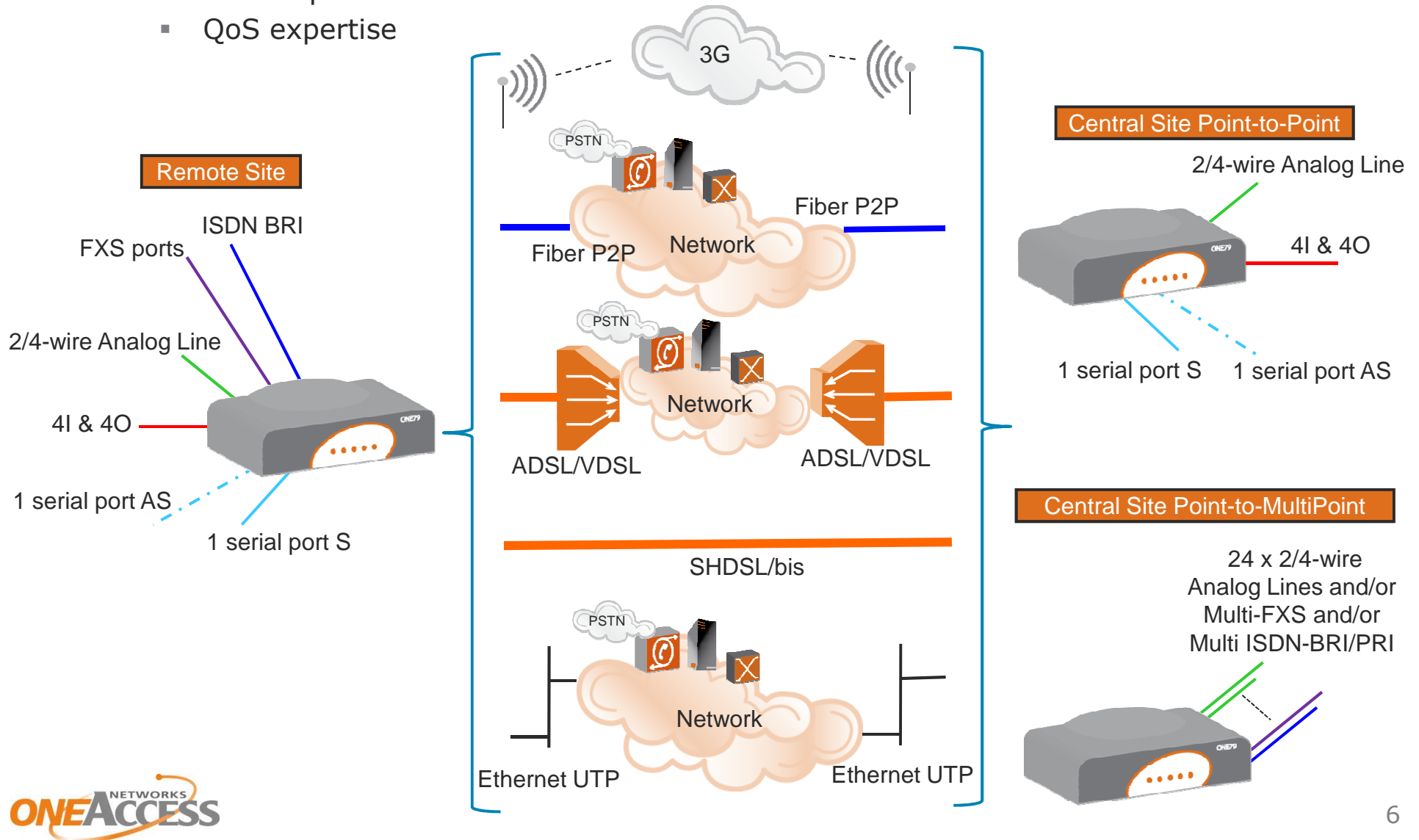
Code	Mnemonic	Description
81302 Swiss- Edition	ONE425-4B-4V AV2 F4TEWn/a	ONE425 Fast Ethernet SFP/UTP WAN, single pair A/VDSL2 over POTS, 4BRI (voice service), 4FXS (voice service), 4 port GbE switch, Wifi 11n



Uplink Mix and User Ports Mix



- OneAccess solutions based on OneOS
 - VoIP expertise
 - QoS expertise

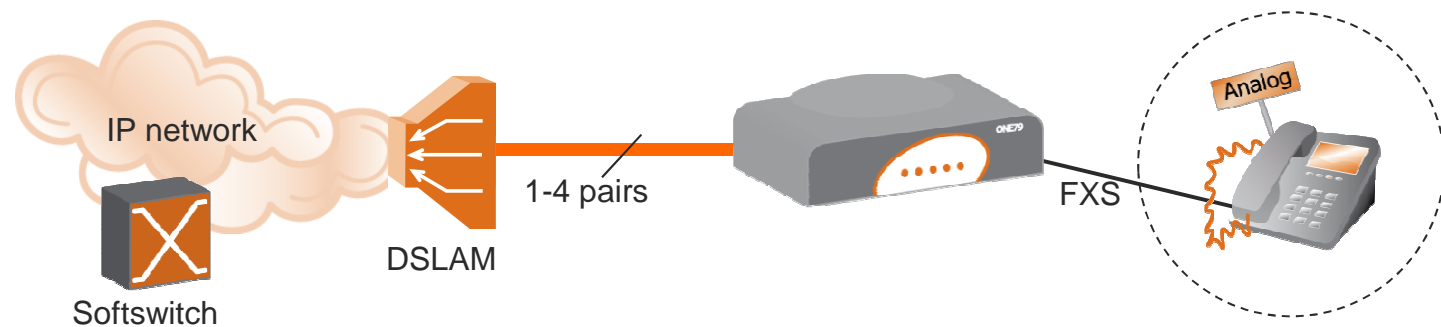




SIP FXS Gateway



- DTMF or Pulse dialing
- Hook Flash (R) detection (Broadsoft compatible)
- Support of CLIP (Caller Id): ETSI/FSK or DTMF
- Configurable tones (dialing, busy, ringback, error)
- Media channel redirection / hold supported
 - Allows support of softswitch services: call transfer, 3-party, music on-hold,..
- Support of # * in dialing phase
 - Enables configuration of services (e.g.: Call Forwarding) directly handled by the softswitch
- FXS compliant with ETSI EN 202 971 V1.2.1



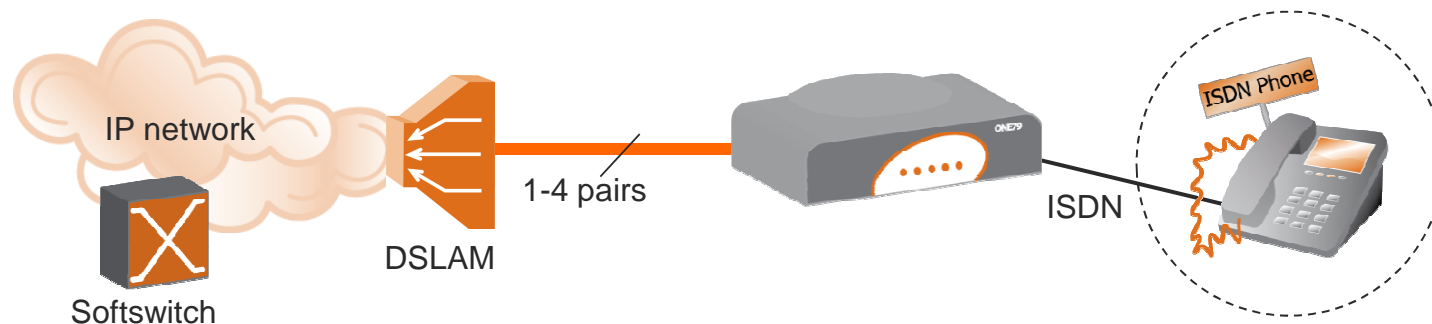


SIP ISDN Gateway



- Reduce operating costs by providing ISDN services over inexpensive technologies
- Migration of ISDN users over SIP VoIP networks
- Signaling: ETSI/Euro-ISDN, Euronumeris, Euronumeris+
- Support of in-band progress indicator
 - ALERT, PROGRESS, RELEASE COMPLETE messages
 - Enables ringback tones or voice messages before or after the call establishment
- Multiple Subscriber Number (MSN) per port, Direct Dial-In (DDI)
- Support of Unrestricted Digital Information (UDI) 64kbit/s

VoIP network sees all ISDN terminals as being native SIP terminals

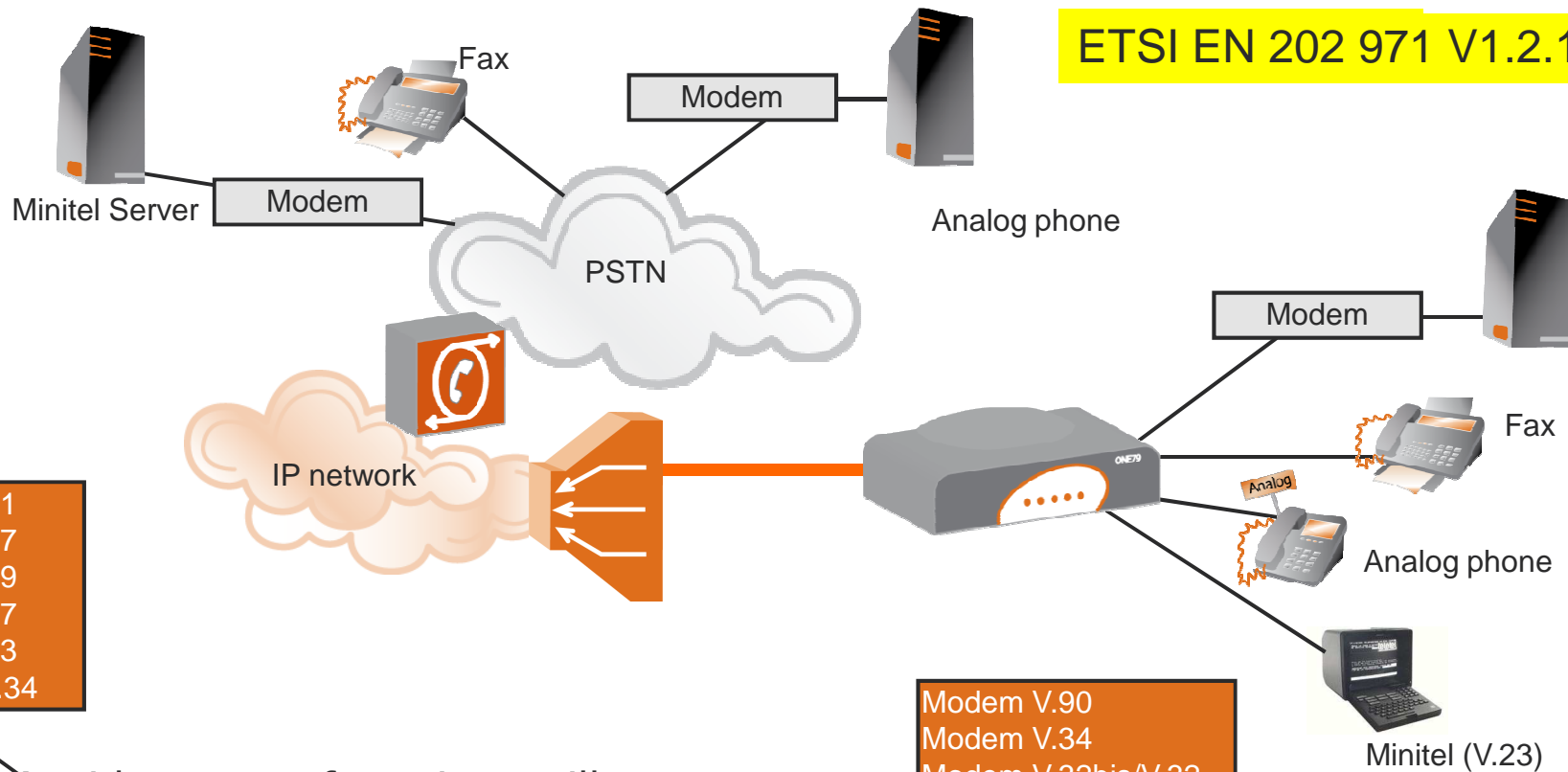




Fax/Modem Support in VoIP



ETSI EN 202 971 V1.2.1



G3 Fax V.21
 G3 Fax V.27
 G3 Fax V.29
 G3 Fax V.17
 G3 Fax V.33
 SG3 Fax V.34

Modem V.90
 Modem V.34
 Modem V.32bis/V.32
 Modem V.22bis/V.22
 Modem V.21 FSK
 Modem V.23

- A wide range of services still uses:
 - Fax
 - Modem ie. Data Protocols
 - Payment terminals (modem)
- Operators and business organizations are moving toward VoIP networks
- OneAccess Solution :
 - Comprehensive support of Fax/Modem over VoIP networks





Features for Modem/Fax Families



CLI configuration	action on CED tone detected (1)	action on ANS or ANSam tone detected (2)
no echo-disable no modem-passthrough no fax-relay passthrough	no action taken	no action taken
echo-disable voicemodem no modem-passthrough no fax-relay passthrough	disable NLP part of echo canceller disable DTMF detection disable VAD	disable echo canceller disable DTMF detection disable VAD
echo-disable modem no modem-passthrough no fax-relay passthrough	disable echo canceller disable DTMF detection disable VAD	disable echo canceller disable DTMF detection disable VAD
modem-passthrough no fax-relay passthrough	disable echo canceller (3) disable DTMF detection (3) disable VAD (3) switch to G.711 coder (3)	disable echo canceller disable DTMF detection disable VAD switch to G.711 coder
fax-relay passthrough no modem-passthrough	disable NLP part of echo canceller (4) or disable echo canceller (4) disable DTMF detection disable VAD switch to G.711 coder	no action taken
fax-relay passthrough modem-passthrough	disable NLP part of echo canceller (4) or disable echo canceller (4) disable DTMF detection disable VAD switch to G.711 coder	disable echo canceller disable DTMF detection disable VAD switch to G.711 coder

(1): CED tone is used by **Group 3 (G3) faxes and low speed modems up to V.29 modulation (9600 bps)**. G3 fax detection is done according to `detection-fax` configuration (V.21 or CED).

(2): ANS or ANSam tone is used by **Super Group 3 (SG3) faxes and high speed modems (from V.32 modulation and above)**

(3): in that case G3 faxes are considered as low speed modems

(4): according to `echo-disable` configuration

VAD : Voice Activity Detection
NLP : Non Linear Processing



SWISS Analog Interface Technical



Telefonanschluss-Kenndaten

Gleichstromschleife

Speisespannung:	48 V (Pluspol an Erde)
U _{App} bei aufgelegtem Mikrotel:	48 V
U _{App} bei abgehobenem Mikrotel:	4 .. 10V
Gleichstrom:	20...50 mA
Schleifenwiderstand:	960...240 Ohm
Speisedrossel-Widerstand:	2 x 350 Ohm
Endgerätewiderstand:	200 Ohm
T _n -Leitungswiderstand:	60 .. 1500 Ohm

Wechselstromschleife

Sprechwechselspannung:	0,2...0,3 V (max. 0,8 V)
Sprechwechselstrom:	0,3...0,5 mA (max. 1,3 mA)
Übertragungsfrequenz:	300...3400 Hz
Schleifenimpedanz:	600 Ohm bei 800 Hz

Rufsignal

Rufspannung: am Endgerät:	70 V ~, überlagert auf die Gleichspannung 15...70 V ~ je nach Leitungslänge
Rufstrom:	20...28 mA
Ruffrequenz:	25 Hz bei Amtszentralen 50 Hz bei TVA's
Rufrythmus:	1 s Ruf 4 s Pause

Summton

Empfangsspannung:	0,38 V (-6dB)
Strom:	0,63 mA
Frequenz:	425 Hz bei Amtszentralen 500 Hz bei TVA's

Besetzton

Wie Summton	
Rhythmus:	1/4s Ruf / 1/4s Pause

Taximpulse

Empfangsspannung:	0,1...2 V
Impulsstrom:	0,5...10 mA
Frequenz:	12kHz ± 1 %
Impulsdauer:	50...70 ms



Compatibility Chart



Requirements	One425
Gleichstromschleife DC Loop	Yes
Wechselstromschleife AC Loop	Yes
Rufsignal Ringing	Yes Ringing Pattern configurable
Summton Dialtone	Yes Dialtone Pattern configurable
Besetztton Buzytone	Yes BusyTone configurable
TaxImpulse	Yes configurable



Fax/Modem Transmission



- T.38 Fax Relay (G3 Fax)
 - Fax Relay applies to Fax G3 transmissions
 - The fax signal is demodulated by the calling gateway
 - to extract the T.30 content
 - to encapsulate it into Internet Fax Packets (IFP).
 - At the remote end the called gateway
 - extract the T.30 content from the IP packets
 - re-generates the analog fax modulation signal
- Fax/Modem Pass through
 - The fax/modem signal is passed in-band through the VoIP network using G.711 coding
- Jitter Buffer to deal with the variations in delay and jitter
 - clock sensitive applications such as fax and modem
 - a jitter buffer is implemented in the IAD at the receiver side
 - Default value: 100 ms
 - directly managed in DSP for accurate real-time processing

Fallback from
SG3 fax to G3
fax

Fallback from
T.38 to Pass-
through



G3 Fax ECM & Redundancy



- Error Correction Mode (ECM)
 - applies to G3 fax
 - allows fax machines to detect fax transmission errors
 - retransmits erroneous frames

- ECM feature over T.38
 - can be turned on and off
 - disabling the ECM feature overrides the fax device capability

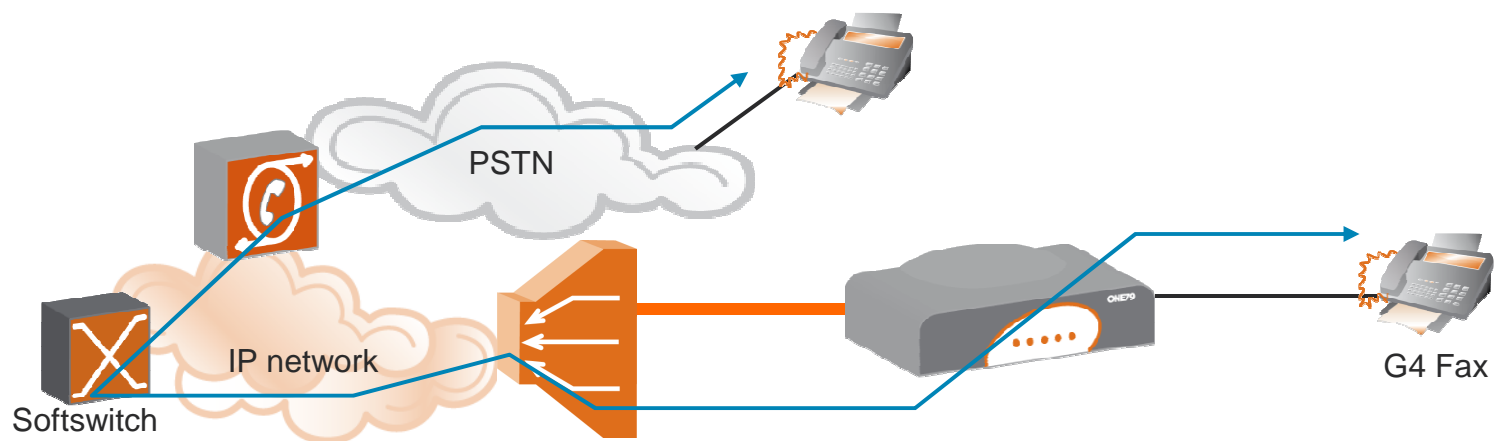
- T.38 redundancy with different levels (from 0 to 4)
 - defines the number of times T.38 UDP packets are repeated.
 - OneAccess recommends to use T.38 redundancy over unreliable networks with packet loss and a lot of jitter



ISDN Fax Transmission



- G4 fax machines
 - designed for use with ISDN networks
 - Unrestricted Digital Information (UDI) 64kbit/s channels
- The calling gateway offers the Clear Mode pseudo-codec for session establishment (SIP-SDP)
- Fax data are carried transparently
 - RTP packets
 - No encoding/decoding





Thank you

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