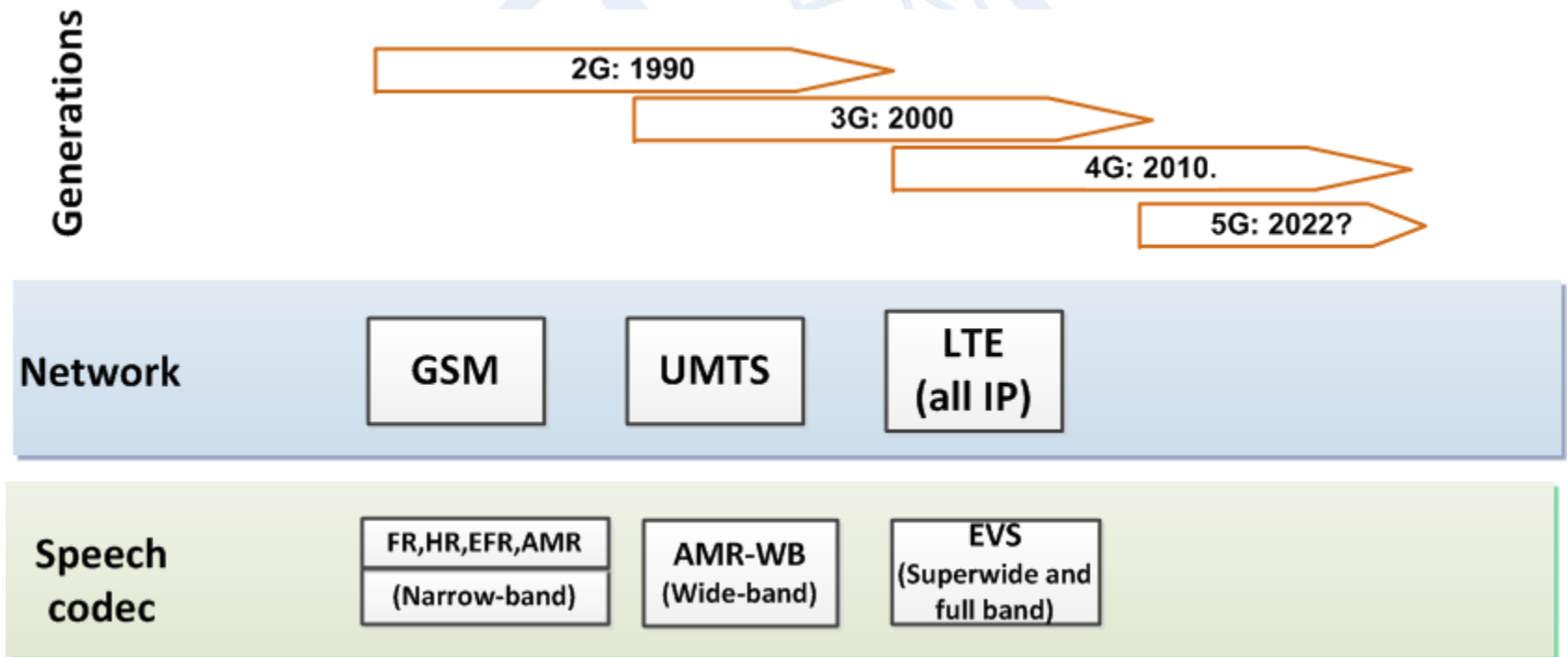

**ITU Workshop on “Voice and Video over LTE”
Geneva, Switzerland, 1 December 2015**

**Enhanced Voice Services (EVS) - Latest state-of-the-art
speech and audio communication codec and related
interoperability aspects**

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Speech Codec evolution in mobile phones



Introduction

- What is EVS?
 - EVS = Enhanced Voice Services
 - The next generation 3GPP communication codec (after AMR-WB, 2001)
 - Substantially improved with respect to
 - Speech quality and compression efficiency
 - Quality for non-speech content (mixed content, music)
 - Audio bandwidth (superwideband, fullband)
 - Error robustness
 - Integrated AMR-WB for seamless switching from/to EVS
 - Result of a cooperation of 12 companies:

ERICSSON 

 HUAWEI

 NTT

 orange™

QUALCOMM

 VoiceAge®

 Fraunhofer
IIS

NOKIA

NTT
Do Co Mo 

Panasonic

SAMSUNG

ZTE中兴

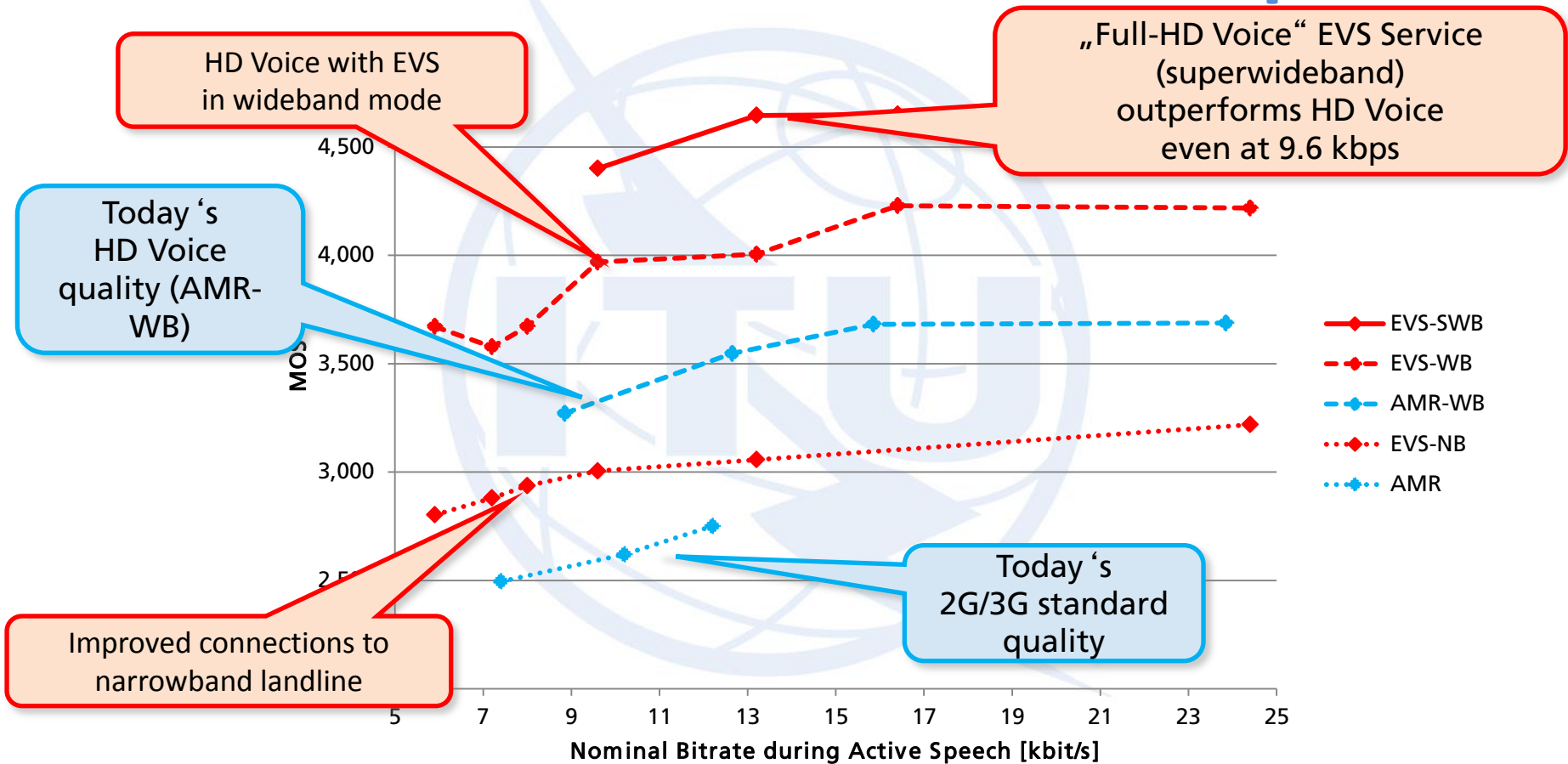
EVS Status

- 3GPP
 - EVS for packet switched (4G) standardized in September 2014
Primary use case is VoLTE, but also fit for VoWiFi, fixed VoIP
 - Extensive performance data available in 3GPP TR 26.952
 - Ongoing work on specifications to enable the use of EVS in Circuit Switched 3G (UTRAN)
- GSMA
 - EVS integrated into VoLTE Specification IR.92 in March 2015
EVS Mandatory for SWB, optional for WB and NB services

EVS Performance

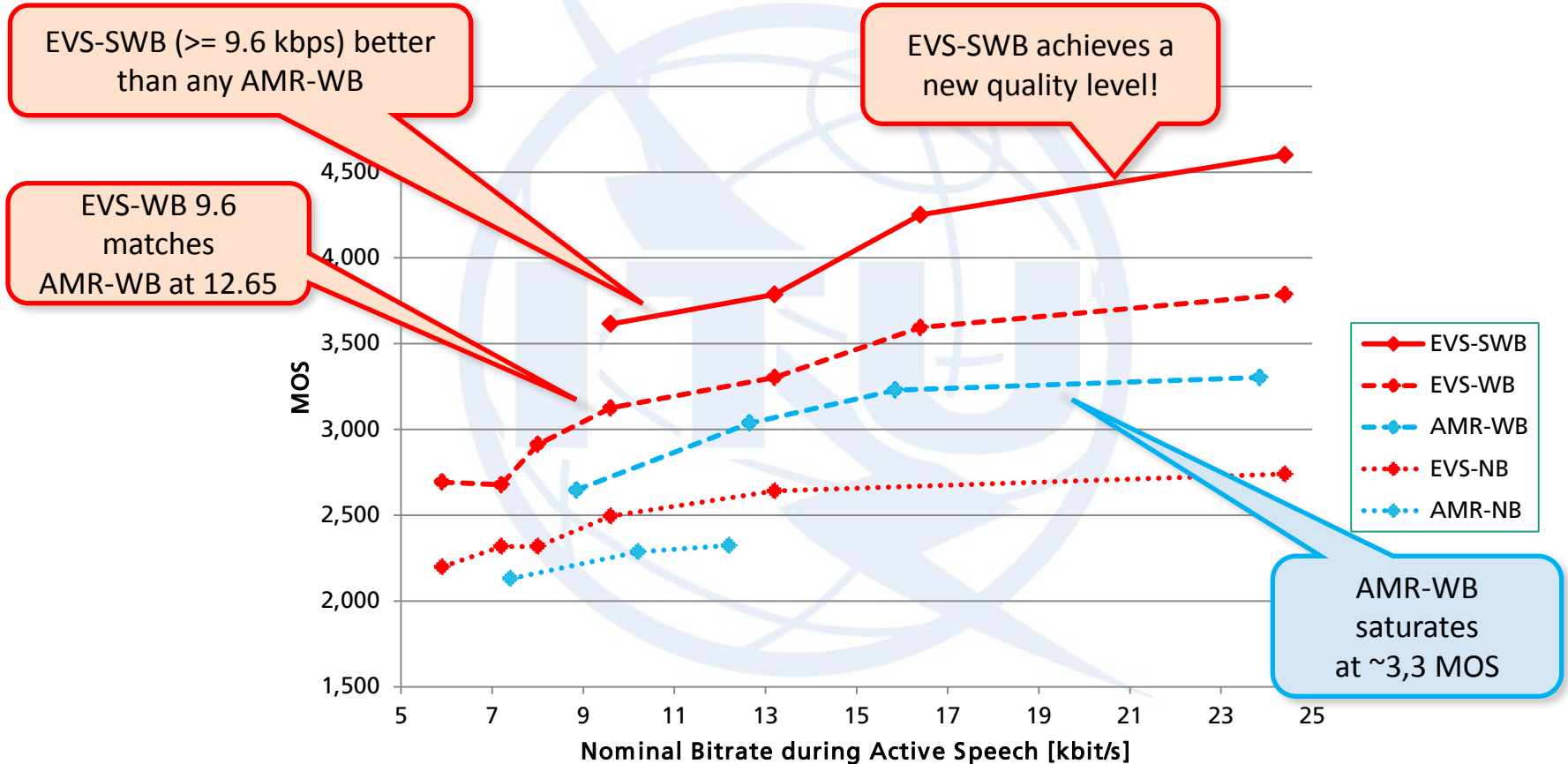


EVS Performance Gain – Clean Speech



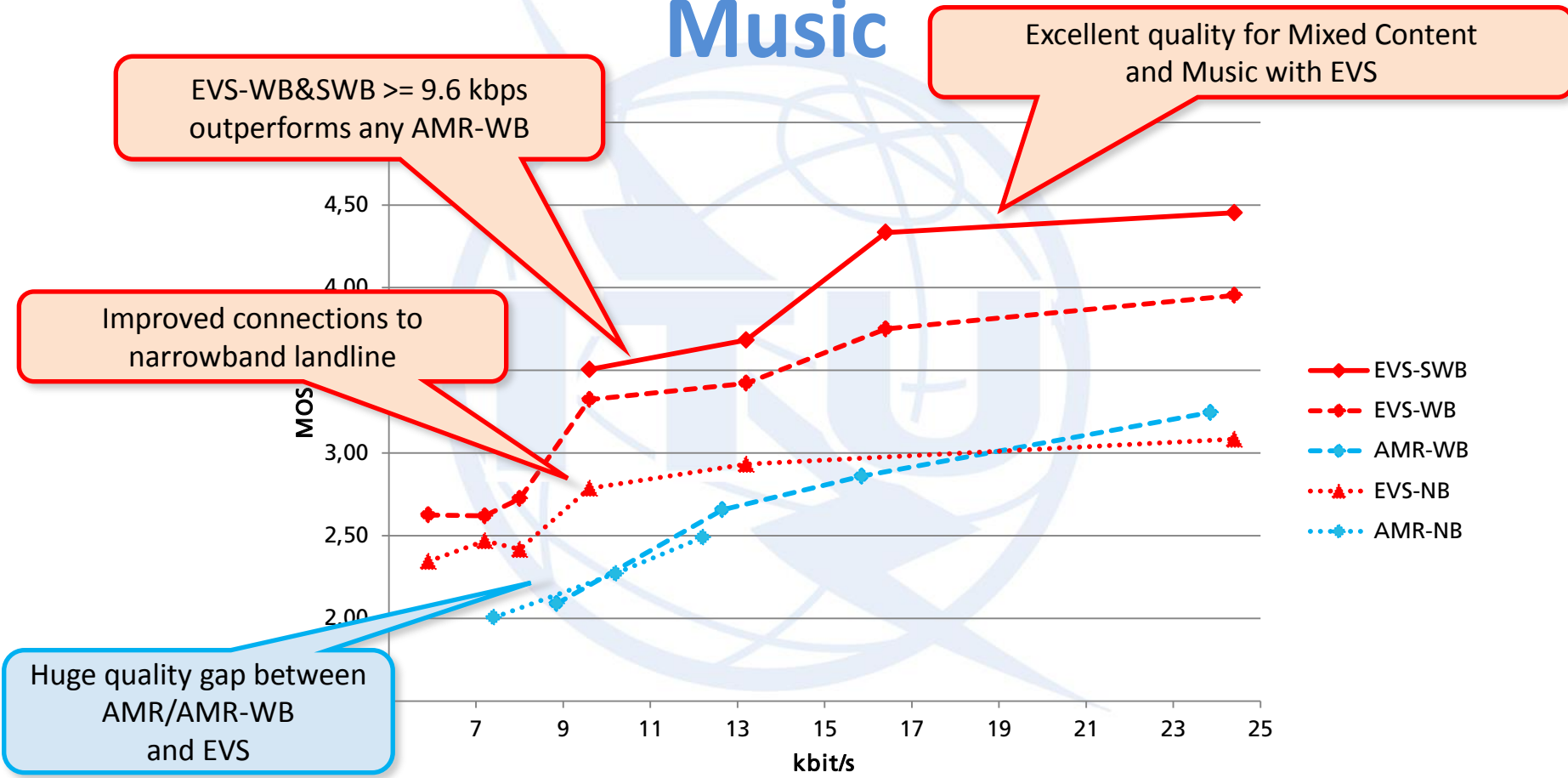
Source: 3GPP TR 26.952, Experiment M1 (mixed bandwidth), Clean Speech, DTX on, North American English

EVS Performance Gain – Noisy Speech



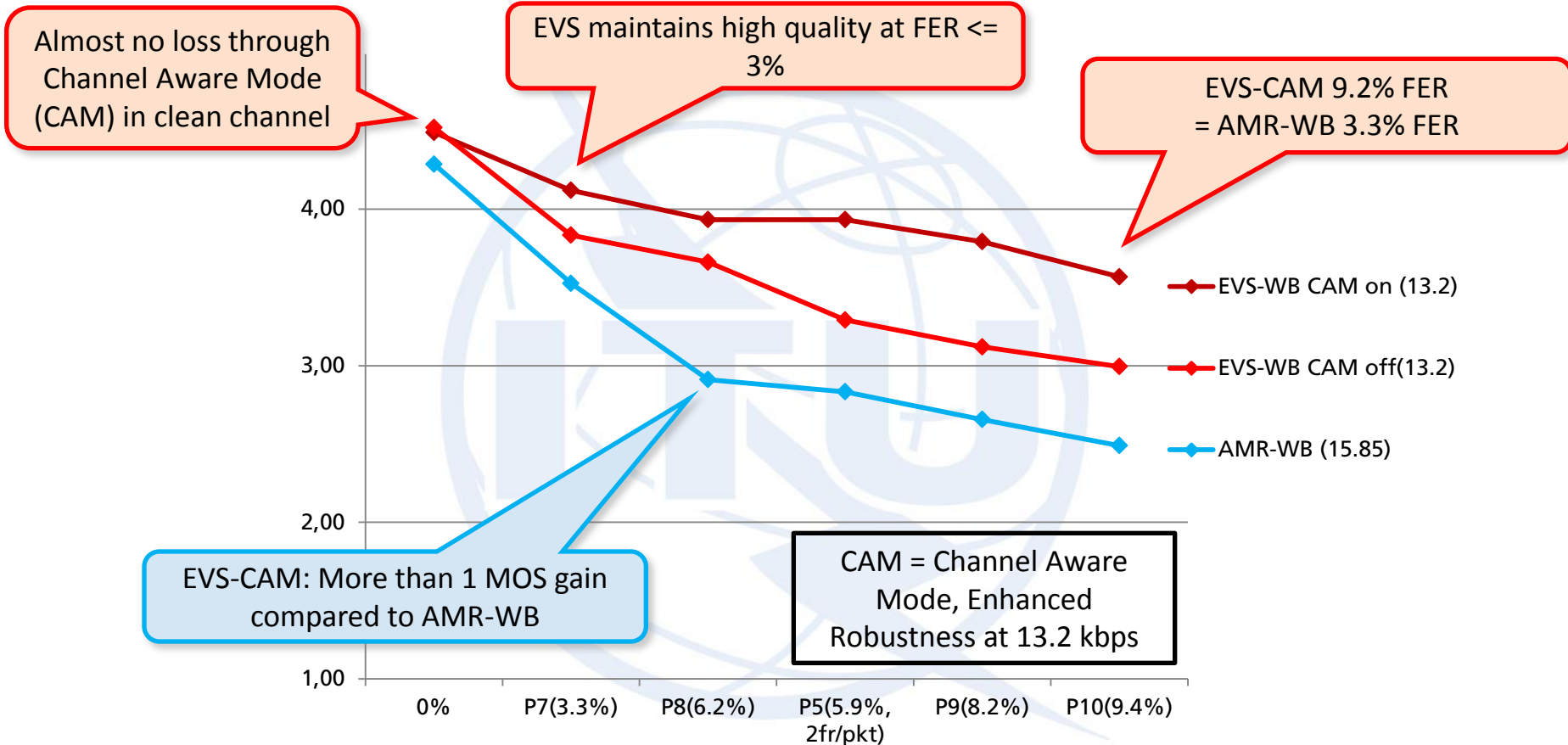
Source: 3GPP TR 26.952, Experiment M2 (mixed bandwidth), Noisy Speech (Car Noise 20dB), DTX on, Finnish

EVS Performance Gain – Mixed and Music



Source: 3GPP TR 26.952, Experiment M3b (mixed bandwidth), Mixed and Music, DTX on, North American English

EVS Performance Gain – Error



Wideband Clean Speech, North American English
 FER=Frame Error Rate, CAM = Channel Aware Mode

Source: 3GPP TR 26.952, Characterization Experiment W1

EVS Performance Summary

- Higher efficiency and transparent quality for wideband and narrowband services
 - Up to transparent wideband speech (at 24 kbps)
 - Up to transparent wideband mixed content and music (at 24 kbps)
 - Substantially improved compression efficiency at all rates
 - High robustness against packet loss – fit for Voice over WLAN
- Unprecedented quality through „Full HD Voice“ superwideband audio at mobile bitrates
 - 14-16 kHz audio bandwidth from as low as 9.6 kbps
 - Highest quality speech, mixed content and music
 - Outperforms wideband at any operation point
- Integrated AMR-WB interoperable mode
 - Improved quality and robustness while 100% compatible with AMR-WB

EVS Interoperability aspects



EVS Interoperability aspects (1)

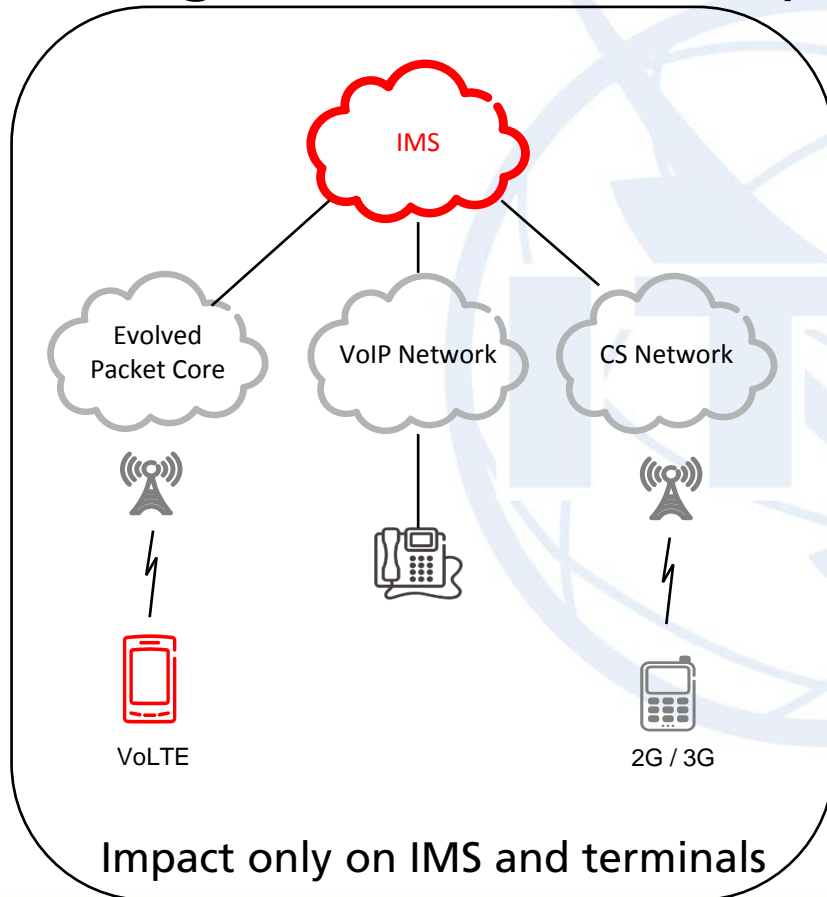
- EVS supports narrow-band, wide-band, super-wideband, and full-band
 - -> Bandwidths for all device classes
- EVS supports bitrates from 5.9VBR to 128kbps
 - -> Rates for a large variety of mobile and fixed networks
- EVS includes fully interoperable AMR-WB encoder and decoder
 - TS 26.114 and IR.92: EVS is alternative implementation for AMR-WB
 - > single codec for AMR-WB and EVS
 - Mode can be changed from EVS to AMR-WB and back within the codec
 - > Handover without transcoding/re-negotiation for SRVCC

EVS Interoperability aspects (2)

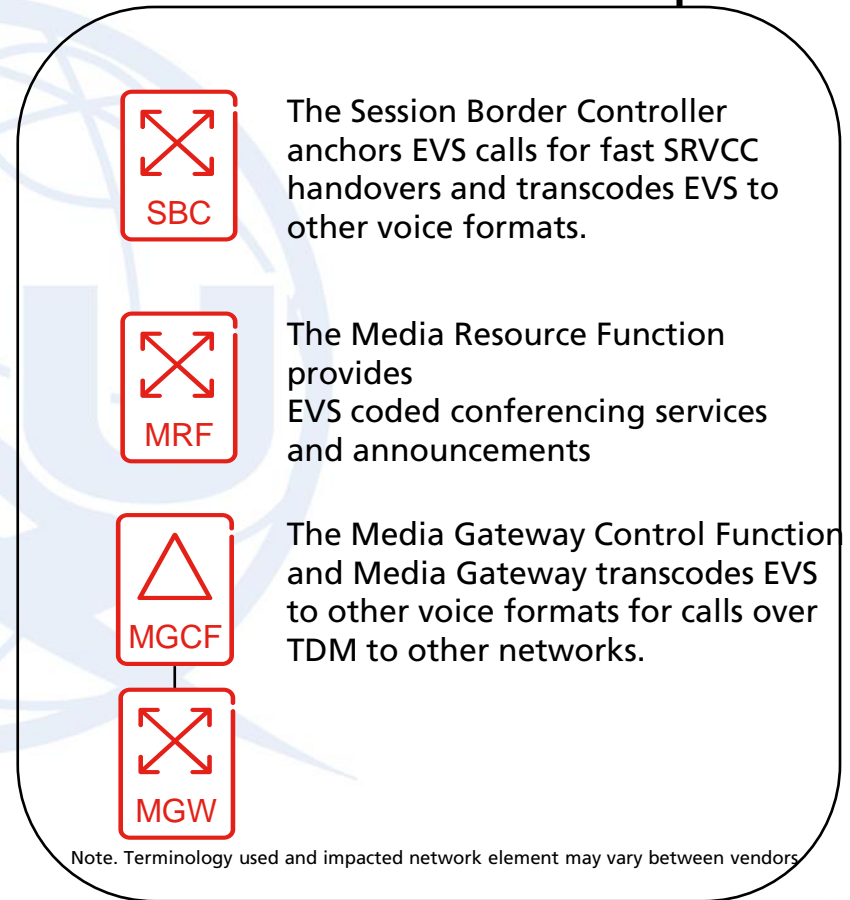
- EVS transport has been designed for LTE, interoperable to AMR-WB
 - Same RTP packet sizes as AMR-WB
 - Constant bitrate
 - VAD/DTX/CNG operation
- > Facilitates easy transition from AMR-WB in VoLTE networks

EVS – VOLTE NETWORK IMPACT

High-level network impact



Detailed IMS impact



Conclusions

- EVS enables operators to offer superior voice services compared to legacy, especially in super wide band mode
- easy integration into AMR-wb optimized VoLTE networks
- Due to its flexibility, EVS can become the „single codec“ for mobile as well as fixed services, including VoWiFi