

MEASURING TRANSMITTER VoIP AUDIO DELAY

Delay verification of VoIP (ED-137B/C) radios with R&S®CMA-K611



Your challenge

New air traffic control (ATC) and MilCom radios transmit and receive audio via VoIP and legacy analog audio interfaces. Radios now need to have their digital VoIP interface tested along with their analog audio interface. The radio needs to have an IP connection that supports the ED-137B/C protocol. VoIP adds another layer to the well-known transmitter and receiver test. Since radios receive audio via IP stream, the audio needs to run through a jitter buffer, be processed and sent to the modulator, which can take up time and cause delays. Transmitter delay is an important factor in evaluating audio quality and network setups.

Our solution

The CMA VoIP option R&S®CMA-K610 provides a VoIP interface for both analyzers and generators. The option seamlessly integrates into the user interface for fast switching between analog and digital audio interfaces.

For transmitter VoIP audio delay measurements, the R&S®CMA-K611 option can provide results without the use of any external equipment. Simply press a button.

Your benefit	Features
Audio delay test	Transmitter test at the touch of a button
Seamless integration	The measurement is integrated into existing VoIP and transmitter tests for ease of use
Quality measurement	High measurement precision and repeatability



Setting the VoIP connection

VoIP Settings

VoIP ☒

DUT URI sip: tx @ 10.121.3.210

CMA URI sip: vcs CMA URI [user info]

PCM Codec A-law

Frequency ID

ID 133.000 Channel Spacing 25.00 kHz

Frequency 133.000000 MHz Apply to Measurement RF ☒

Setting up the URI of the radio and CMA

VoIP transmitter delay test

TX Test

Generator VoIP ☒

FID 58.575

URI sip: TxSDTR @ 172.29.20.32

Frequency 58.575000 MHz

Expected Power 30.00 dBm

Method TX Audio Delay

AF Source VoIP

	Current	Average	Maximum	Std. Dev.	Unit
Delay	22.0	18.5	26.6	0.0	ms

Tx Audio Delay

Running

Delay 22.0 ms

Cancel

The CMA provides a signal with a test pattern to the DUT and measures the VoIP transmitter delay.

VoIP transmitter test

TX Test

Generator VoIP ☒

FID 133.000

URI sip: tx @ 10.121.3.210

Frequency 133.000000 MHz

Expected Power 53.00 dBm

Method SingleTone

AF Source VoIP

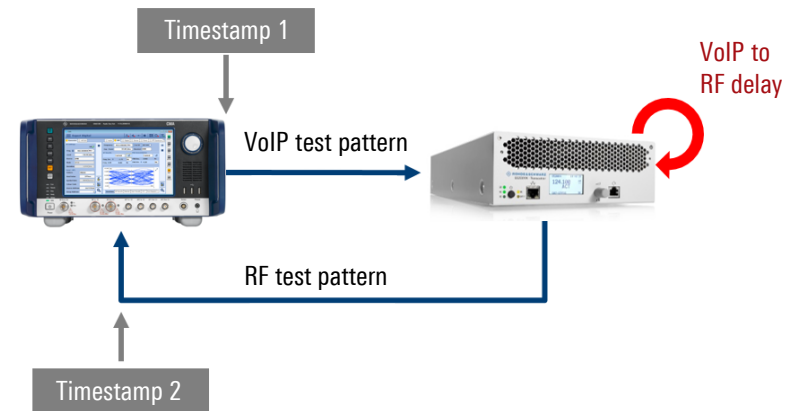
	Current	Average	Maximum	Std. Dev.	Unit
Frequency	1.0208	1.0200	1.0209	0.0006	kHz

Signal Quality

	Current	Average	Extreme	Std. Dev.	Unit
THD[%]	0.0587	0.0592	0.0628	0.0011	%
THD[dB]	-64.63	-64.55	-64.04	0.16	dB
THD+N	0.9774	0.9779	0.9849	0.0009	%
SINAD	40.20	40.19	40.22	0.01	dB
S/N	40.21	40.21	40.23	0.01	dB
(S+N)/N	39.13	39.13	39.15	0.01	dB
(S+N+D)/N	39.13	39.13	39.15	0.01	dB

The VoIP audio generator provides an IP connection and audio input to the radio. The CMA receives the modulated RF signal and measures the RF parameters and signal quality.

Setup for VoIP transmitter delay



The VoIP delay is determined as the difference between two timestamps. One when the test pattern leaves the R&S®CMA and a second one when the test pattern is detected by the CMA via RF.