NAME
opusrtp – encapsulate Opus audio in RTP

SYNOPSIS
opusrtp [**-hV**] [**--sniff**] **file.opus** [**file2.opus ...**]

DESCRIPTION
opusrtp Demonstration tool for sending and receiving Opus audio data in RTP, used for interactive applications on the internet.

By default, Opus audio from each given file is sent as an RTP stream.

OPTIONS
-h, --help

Print help message

-V, --version

Display version information

--quiet   Suppresses program output

--sniff   Sniff the network for active RTP sessions and save them to .opus files. This can be useful for debugging other Opus RTP implementations. For this function to work, the program must be run with superuser privileges.

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BUGS
Only the sniff mode is implemented; the tool should do normal unicast and multicast send/receive.

The sniff mode should allow specifying device/host/port/payload type to limit capture. All that is hardcoded.

SEE ALSO
opusdec(1), **opusenc**(1), **opusinfo**(1)