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## What's next for Bluetooth in PulseAudio?

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What is PulseAudio?

- Initial release on 17 July 2004
- First appeared for users in Fedora Linux with version 8
- Mediates access to audio resources on your system
- ► Features?
  - Audio mixing
  - Per application volume controls
  - Multiple sources and sinks
  - Combine multiple sound cards and synchronize multiple playback streams
  - Bluetooth support
  - Command line interface with scripting capabilities
  - Sound daemon with reconfiguration capabilities

Bluetooth Profiles

- ▶ What is a profile?
- Some of the widely used bluetooth profiles
  - Advanced Audio Distribution Profile (A2DP)
  - Audio/Video Remote Control Profile (AVRCP)
  - Headset Profile (HSP)
  - ► Hands-Free Profile (HFP)

**Bluetooth Codecs** 

- Available codecs?
  - ► Low-complexity subband codec (SBC)
  - Audio Processing Technology (aptX, aptX-HD)
  - Advanced Audio Coding (AAC)
  - ► LDAC

So which is better? asymptotic

HOW STANDARDS PROLIFERATE: (SEE: A/C CHARGERS, CHARACTER ENCODINGS, INSTANT MESSAGING, ETC.) 500N: 14?! RIDICULOUS! WE NEED TO DEVELOP ONE UNIVERSAL STANDARD SITUATION: SITUATION: THAT COVERS EVERYONE'S THERE ARE THERE ARE USE CASES. YEAH! 14 COMPETING 15 COMPETING STANDARDS. STANDARDS.

Figure 1: XKCD - Standards

Codec Latencies!

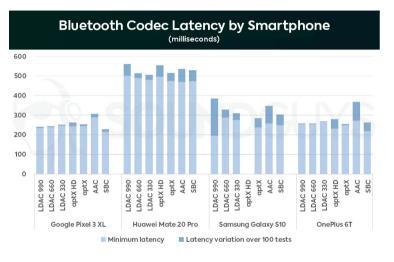


Figure 2: Soundguys - Android Bluetooth Latency

Current State asymptotic

- ► Upstream only supports SBC :(
- ► HSP/HFP support
- Community efforts: out-of-tree pulseaudio-modules-bt, another implementation submitted upstream
- Messaging API

Challenges

- Codec switching
- ► Patent encumbered licenses?
- ► How to support multiple encoders or decoders?
- ► Contributors? Maintainers?

GStreamer asymptrtic

- ► What is GStreamer?
- ► Why?

```
gst-launch-1.0 -v audiotestsrc !
audio/x-raw,rate=44100,channels=2,format=S32LE !
ldacenc eqmid=2 ! a2dpsink
transport=/org/bluez/hci0/dev_4C_BC_98_80_01_9B/sep10/fd0
```



Progress so far

- ► LDAC support upstreamed in GStreamer
- ▶ Support for Low-Overhead MPEG-4 Audio Transport Multiplex (LATM) AAC in GStreamer
- ► GStreamer wrapper around *libopenaptx*. Work done by Igor Kovalenko.
- ► Implementing codec switching
- Proof of concept with GStreamer tested
- ► Merge request opened in PulseAudio

Questions? asymptatic

- ► So what's next?
  - Support Adaptive Bit Rate (ABR) in LDAC
  - mSBC support for HFP profile
  - LC3 codec
  - Allowing users to specify the preference order for codecs
  - Support Opus as a vendor codec for PulseAudio <-> PulseAudio