# **A New RTSP Source Element** Written in Rust

### rtspsrc2

**GStreamer Conference 2024** 

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## Who Am I? **Platform Stuff, Rust, WebRTC, RTSP, Meson, Cerbero, Cl** Whatever needs doing

Everywhere and also Nowhere





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- Some of these have workarounds that every app must implement, others are unfixable







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  - Not tenable in rtspsrc
  - Workaround: add some sleeps in the application <?>













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- End-Of-Life









Sovereign Tech Fund



GStreamer

Re-implementing a multimedia framework's low-level network protocol for increased security, maintainability, and sustainability





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Technologies

Sovereign Tech Fund

"GStreamer RTP sessions in Rust" by Matthew Waters

- RTP send and recv bins
- rtpjitterbuffer
- payloaders, depayloaders
- New RTSP source element

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### GStreamer

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## rtspsrc2: Features

- Live streaming N audio and N video, with RTCP-based A/V sync
- Lower transports: TCP, UDP, UDP-Multicast
- RTP, RTCP SR, RTCP RR
- OPTIONS DESCRIBE SETUP PLAY TEARDOWN
- Supports both the old rtpbin and the new rtpbin2 rust rewrite Set USE\_RTPBIN2=1 to use rtpbin2
- Lower transport selection **and** priority (NEW!)
  - Even supports different transport for each SETUP call
  - F.ex., KLV could be sent over TCP and the rest over UDP
- port-start instead of port-range







## rtspsrc2: Characteristics

- rtsp-types and sdp-types crates
- rtpbin2  $\Rightarrow$  all reading, parsing, writing, etc is done in Rust
  - Except codec parsing and demuxing
- Async Rust, multi-threaded Tokio
  - Simpler and decoupled, easy to follow
- Network I/O
  - Does not use udpsrc, udpsink
  - Can be made pluggable or even I/O-less







## rtspsrc2: Characteristics

- appsink + appsrc + ghostpads
  - Simplified event handling, startup/shutdown
  - Expose pads early, sparse streams
- Performance on-par with rtspsrc
- Tested against live555 and gst-rtsp-server
- Most features have not been implemented yet
- Plan is to replace rtspsrc









# rtspsrc2: Low-Hanging Fruit

- Credentials support
- TCP over TLS support
- NAT hole punching
- SET\_PARAMETER & GET\_PARAMETER support
- Option to reconnect TCP control on disconnect
- Missing configuration properties:
  - latency & do-rtx
  - do-rtcp
  - iface
  - user-agent







## rtspsrc2: Giraffe Food

- Stream Collection API
  - Live streams first, VOD later
- SRTP support, incl. all key exchange mechanisms Parsing more SDP attributes
- Control and media flow tunnelling over HTTP
- HTTP and SOCKS proxy support
- Make TCP connections optional when using UDP transports
- Explicit media clock sync support, such as RFC7273







## rtspsrc2: Re-Forestation

- RTSP 2.0 support
  - PLAY NOTIFY, Media-Properties, End-Of-Stream, Scale-Change, etc.
  - Needs an accompanying RTSP 2.0 server implementation
- Video-On-Demand support, both RTSP 1.0 and 2.0
  - PAUSE, Seeking, Looping
- ONVIF backchannel support
- ONVIF trick modes support

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It's a fun project for me, and everyone benefits. Please get in touch! nirbheek@centricular.com

Thanks!!



