

Integrating Video Conferencing into Everyday Applications

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Calls integrated

- Calls in their own app
 - Easy!
 - Skype, Ekiga, WLM, etc
- Call directly in other apps
 - NEW!
 - EXCITING!
 - INNOVATIVE!

Ingredients

- Call signalling
 - Telepathy
- Media Framework
 - GStreamer
- Call protocol abstraction
 - Farsight 2

Farsight 2: Overview

- Audio & video conferencing abstraction
- Multiple protocols abstracted
 - RTP (XMPP Jingle, SIP, MSN SIP, etc)
 - MSN Webcam
 - Raw (soon)
 - Others...
- Regular GStreamer element

Widely deployed

- Maemo
- MeeGo
- GNOME's Empathy
- New KDE Telepathy Call UI
- Pidgin
- aMSN

decodebin for calls

- Encoder
- Decoder
- Payloader
- Depayloader
- Network sources
- Network sinks

High level objects

- Participants
 - People
- Sessions
 - Media types (audio, video)
- Stream
 - Intersection of Participant and Session

Session

- One type of media (audio, video, etc)
- One local media source
 - One microphone
 - One camera
 - File
 - etc
- Multiple stream from other participants
- RTP session

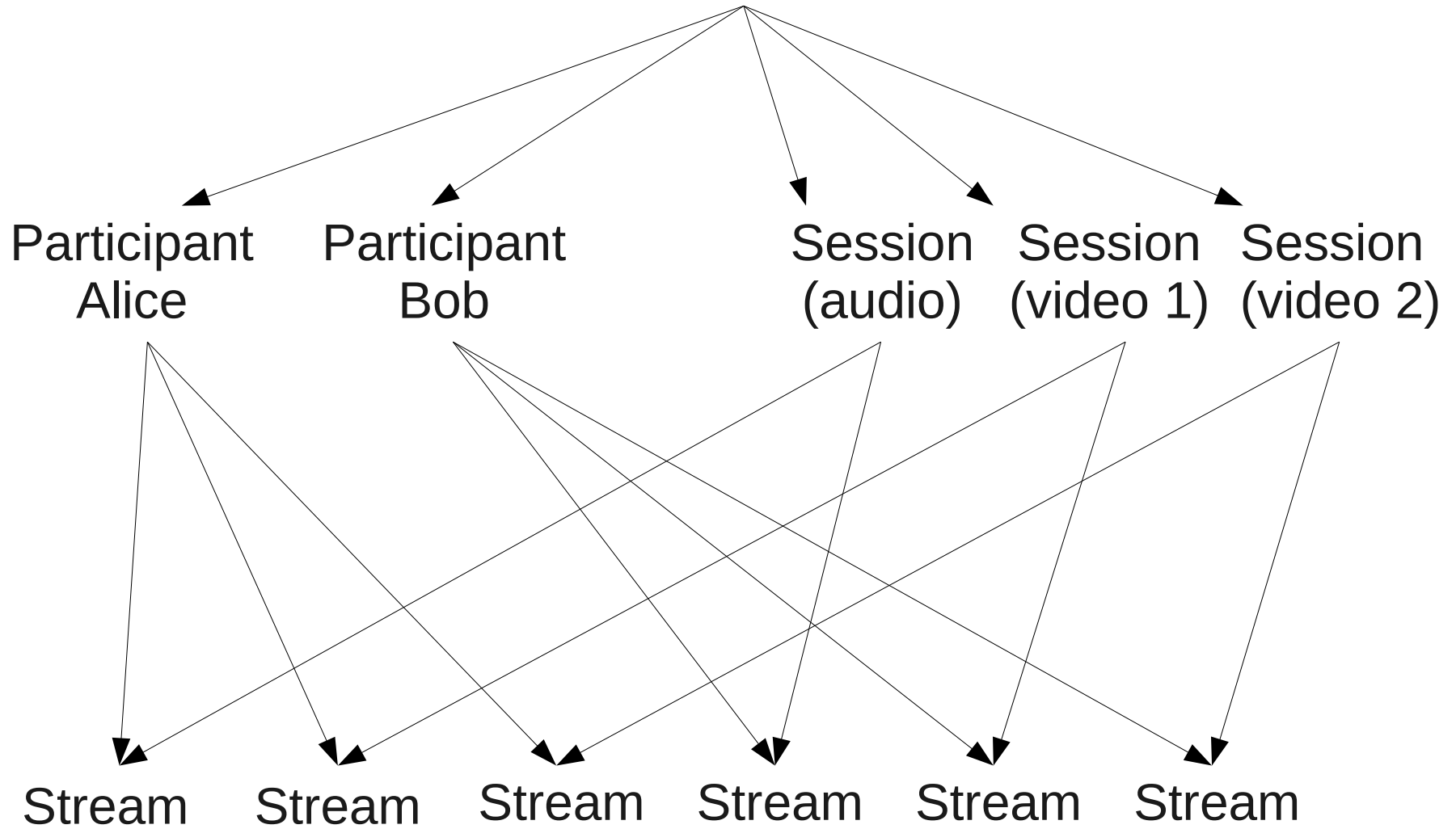
Stream

- One participant in one session
- Use for communication with participant
 - Codecs
 - Candidates
- Remote media comes out of here

Conference

- The GStreamer element
- Multiple synchronized sessions
- Contains everything else

Conference



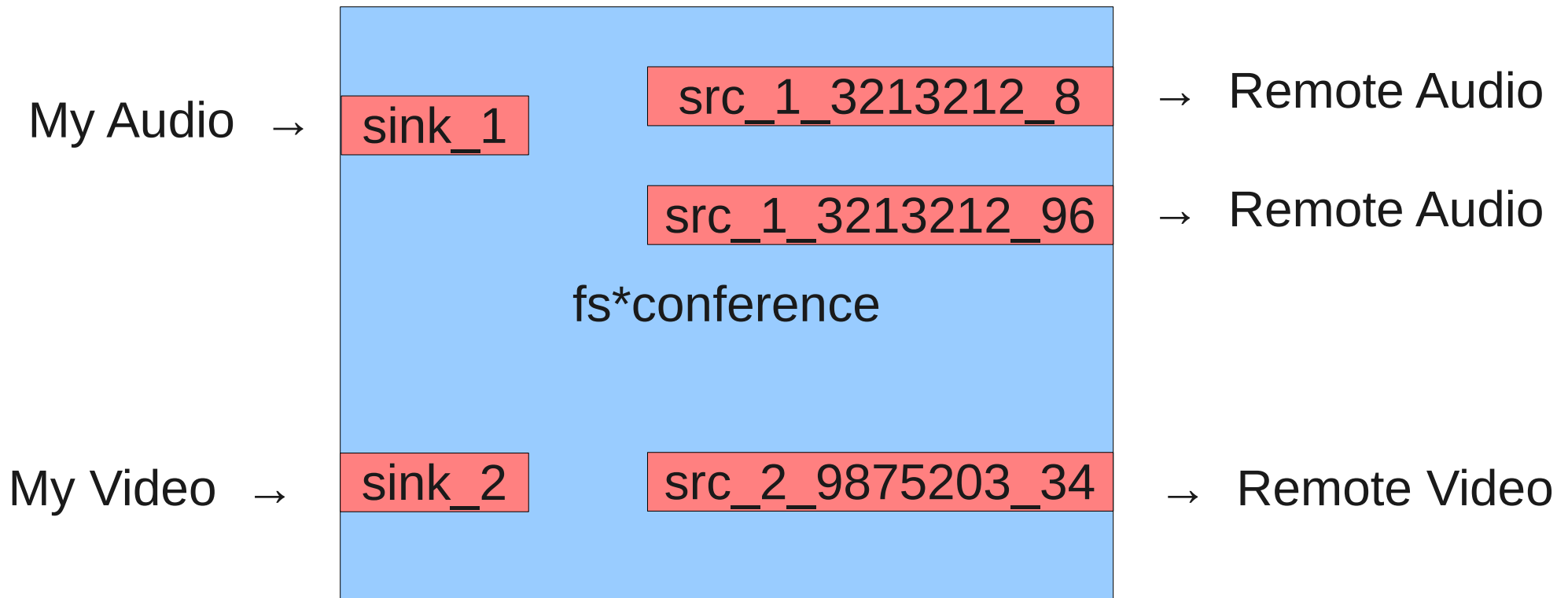
Farsight 2: RTP

- Complete RTP with RTCP
 - Including A/V sync
- Full DTMF
- Offer/Answer Codec Negotiation
- Multiple transports for RTP
 - ICE (RFC, Google and MSN)
 - Unicast UDP
 - Multicast UDP
 - Shared memory

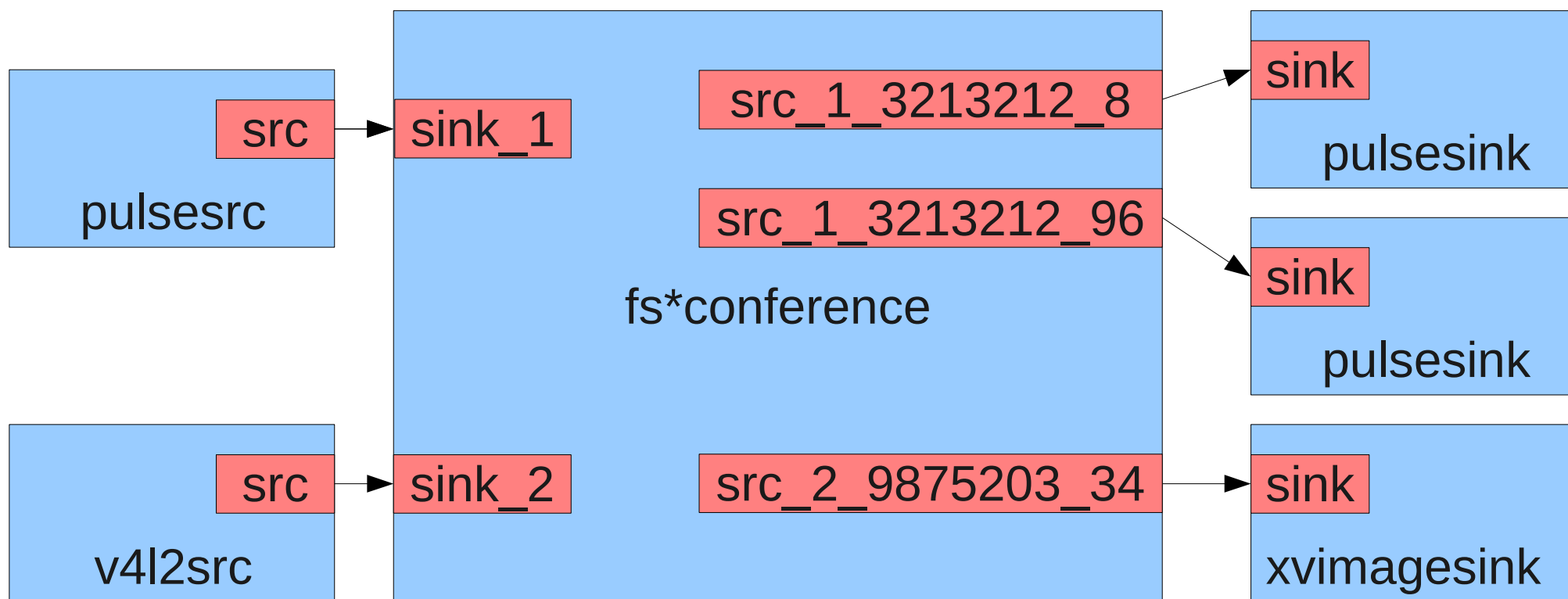
Telepathy Farsight integration

- Telepathy-Farsight library
- Hides all signalling
 - Easy API
- Telepathy streamed media interface limited
 - Only one-to-one
 - Strange session/stream separation
- New Call interface is the bright future
 - Will require some improvements in tp-fs

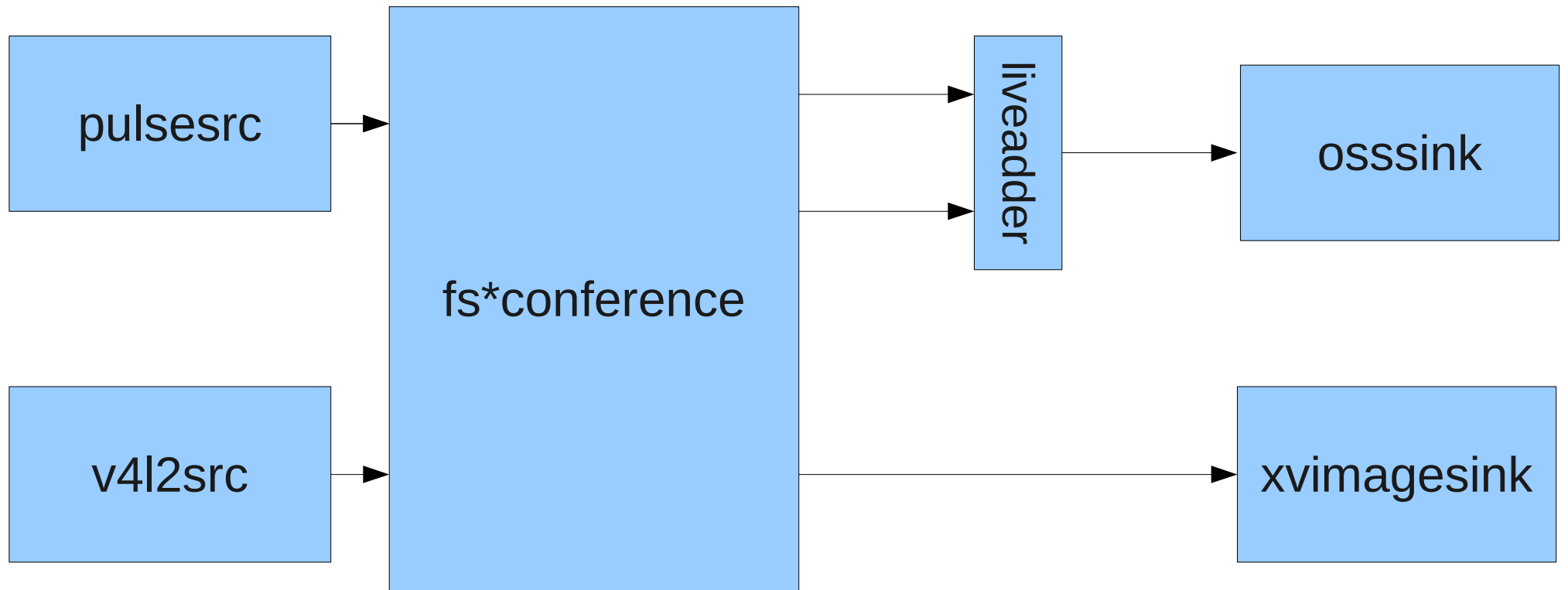
Farsight 2 element



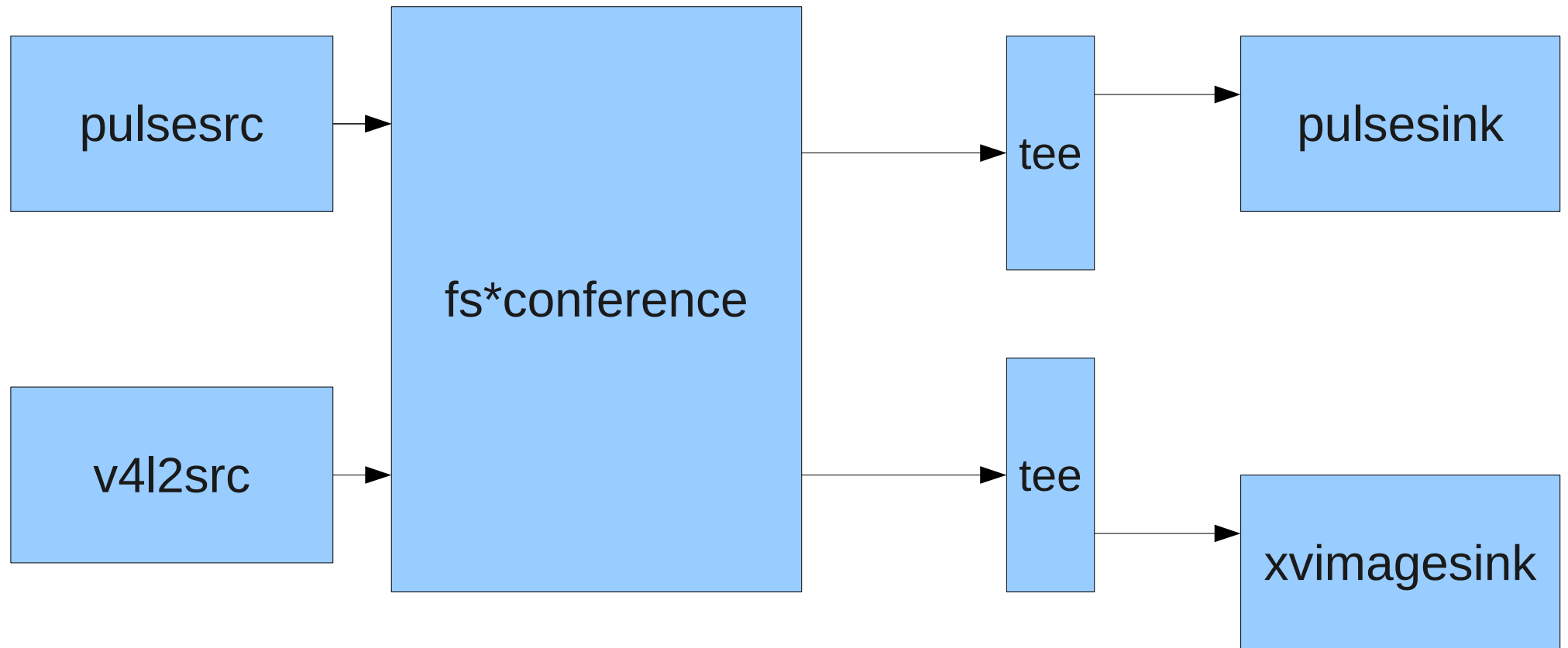
Integrated Farsight 2: Simple Case



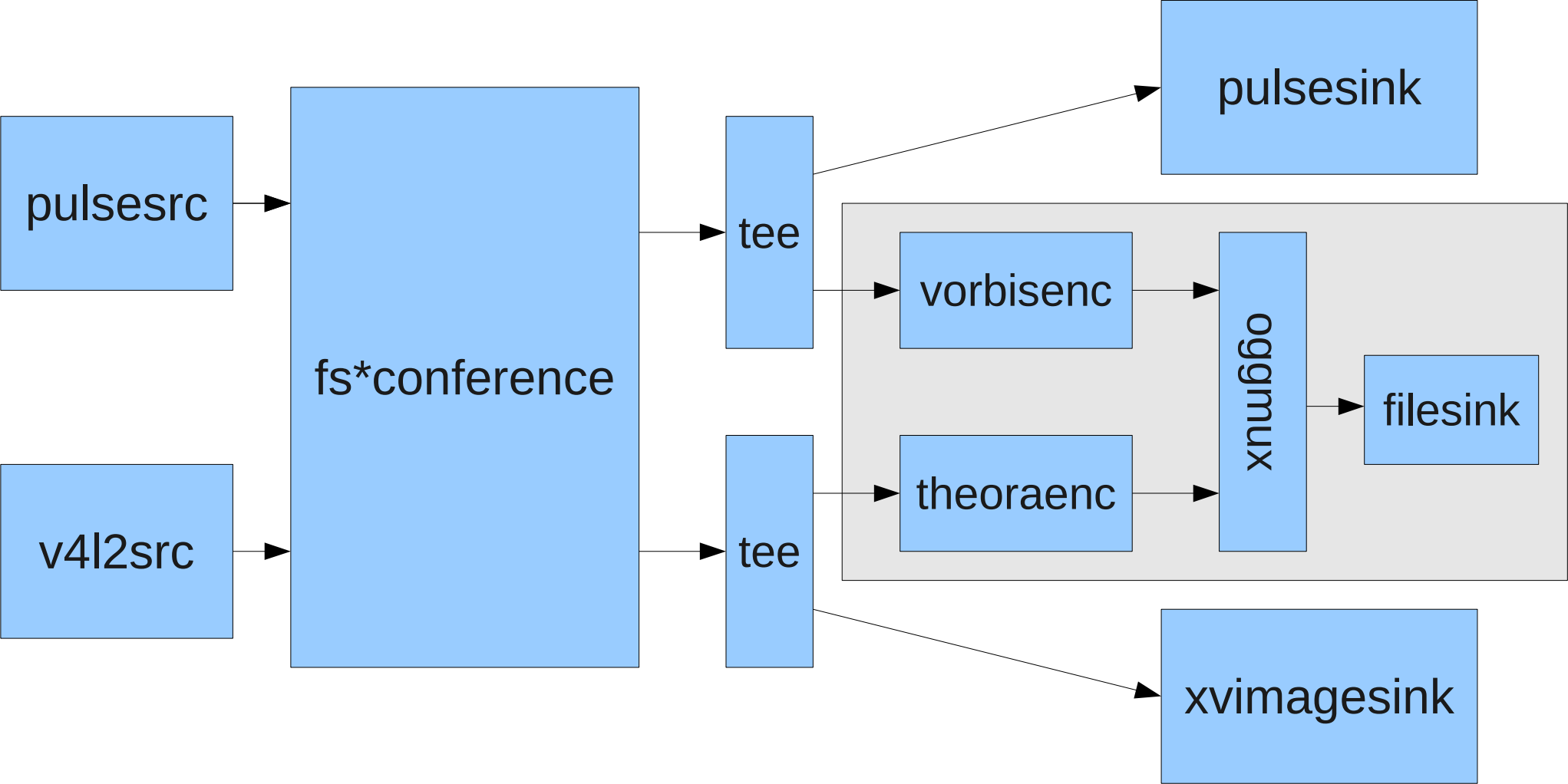
Another simple case



Adding More stuff



Add recording



Steps to get a Telepathy call

1. Get D-Bus bus
2. Get Telepathy Account & Contact
3. Find the handle for that contact
4. Request a Telepathy Channel
5. Wrap it into Telepathy-Python
6. Wrap it into Telepathy-Farsight
7. Attach session-created and session-created
8. Add members
9. Request streams

Steps to get a Telepathy call

10. Put the conference into the pipeline on Telepathy session creation
11. Add/Link the sink/source on stream creation

Extras

- Hook up UI to add added streaming-side features
- Example:
 - The Record button adds the recording pipeline

Soon in Farsight2

- Developer ease of use
- Adaptive behaviour for better quality

Ease of use

- Brought to you by Youness Alaoui
- Magical source and sink
- Filters
- Even higher level APIs
 - For custom Uis
 - Adjustable level of abstraction

Auto Sources & Sinks

- Auto-discovery across APIs
 - Video4Linux 1
 - Video4Linux 2
 - DV cams ?
 - Etc
- Dynamically switch
- GstPropertyProbe sucks
 - Need a better Interface

Source

- Adds tee
- Only start source when needed
 - When something is connected (to the tee)
 - Property to disable it (for Hold)

Sink

- Add mixing only if required
- Drop all buffers using easy API (for Hold)

Filter

- Between two pads
- NOT an element, maybe includes many
- Attach to a pad
 - Returns a new pad
- Transforms GstMessages to signals
- Works in both directions

Filter Manager

- Handle adding multiple filters
- Works with playing pipeline
 - Does pad blocking magic
- Simple API

Multi Filter Manager

- Same API as regular Filter Manager
- Same filters
- Multiple pads
- Useful when using sinks that mix

Simple Farsight \Rightarrow Fsu

- FsuConference
- FsuSession
- FsuStream

- Do all the plugging
- Zero GStreamer knowledge required

Telepathy-Fsu

- Library
- Above Fsu & Telepathy-Farsight
- Telepathy Call handler without a UI
- Ideal for Mobile

Adaptive behaviours

- Standards based
- Experimental !!!
- Extended RTP Profile for Real-time Transport Control Protocol (RTCP)-Based Feedback (AVPF)
 - Keyframes on demand
 - NACKs
 - Other feedback
- TCP Friendly Rate Control (TFRC)

More to come ?

- Generic Forward Error Correction (FEC)
 - Which type to use ?
- Interactive Connectivity Establishment (ICE) with SIP
 - SIP Forking & ICE
 - Contiguous ports for fallback
- Secure RTP (SRTP)

What can you do ?

- Innovative Call UIs
- Collaborative text editing
 - OOo, Abiword, Gnumeric, KOffice
- Stream slides with Evince, Okular
- Remote desktop viewing (exists using VNC)
- Mobile UIs
- Other ideas?

Thank you

- #farsight, #telepathy, #gstreamer @ FreeNode
- #empathy @ GimpNet
- <http://farsight.freedesktop.org/>
- <http://telepathy.freedesktop.org/>
- <http://gstreamer.net/>