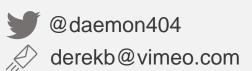
Every Solution is Wrong:

Normalizing Ambiguous, Broken, and Pants-on-Head Crazy Media

Derek Buitenhuis





New phone, who dis?

- Senior Video Engineer @ Vimeo
 - Some things I've worked on:
 - Transcoding pipeline (pre- and post-chunking)
 - Edge stateless segmenting (DASH, CMAF, HLS, etc.)
 - On-the-fly image recompression
 - Captions stack
- Open source developer (FFmpeg, FFMS2, etc.)
- VideoLAN non-profit board member
- Professional Twitter Sh*tposter





Users Gonna User

- Comprehensive ingest guidelines followed by majority of users
 - Important to have; most users/clients will follow these if present
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- Comprehensive ingest guidelines followed by majority of users
 - Important to have; most users/clients will follow these if present
 - The users that don't will send you massively varying media
- We don't have the luxury of demanding they upload correct/perfect media
- We need to be able to ingest this vast array of media as best possible
 - Must be consistent
 - Result must be widely playable while best maintaining the user's intents
 - Anger the least amount of users
- Provide users with a easy to digest recommendations based our analysis

Garbage In, Sanitized Garbage Out

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- Archival purists, please look away now!
- We need to do analysis and run heuristics before transcoding to make informed choices:
 - Does this file have enough well distributed RAPs to efficiently chunk? Can we even seek?
 - Do we need to convert colorspace, and if so, to what? What about HDR?
 - Do we need to scale (SAR/DAR), and to what? Cropping?
 - Do we need to un-screw timestamps, and how? Is there concept of a frame or field rate?
 - Is the file interlaced? Telecined? Is it tagged as such?
 - Do we need to resample audio? Downmix? How?
 - How should we sync audio? Do we need to pad? Silence-fill?
 - Other misc stuff like Apple "Slow-mo", spherical video, MVC, etc.
 - More, more, more...

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- Video:
 - Derive a per-stream score based on various factors
 - Is it marked as a thumbnail stream? Is it marked as default / on?
 - Some media such as slideshows only have a timed thumbnail stream
 - Is it (M)JPEG?
 - Total duration?
 - Bitrate? (taking codec into account!)

- Audio:
 - If downmixing, prefer the official downmixed version, if present, over downmixing ourselves
 - Prefer streams with earlier start times
 - If they all start at the same time, prefer longer durations
 - If everything else fails, go by lowest index

Indexing and you

- Every file / stream is indexed, info collected includes frame types, timestamps, etc.
 - The file is not decoded during this phase, only demuxed in memory
 - Can pass around and store info needed for analysis
 - Seek easily for containers that may not have indexes

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 - Seek easily for containers that may not have indexes
- Much faster than a full decode, especially for things like JPEG2000
- Essentially building a packet-to-frame mapping
 - Harder than it sounds due to things like:
 - alt-refs VP8 requires packet inspection, VP9 may, AV1 doesn't
 - NVOPs Need to be skipped.
 - PAFF Need to handle field packets
 - Virtual timelines (edit lists, ordered chapters)

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- Can analyse how reasonable seeking will be based on frame types
 - We don't want to try seeking in files with one or no keyframes (no block-level ref analysis)
 - Something like a weighted quantile in the time domain based on keyframe distances (90th percentile)
 - How likely are we to end up with a RAP within N frames of a seek?
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- Preferable to chunk based on shot (or rather, cost effective) boundaries, if available
 - Note that source RAP placement is not necessarily indicative of a Good™ chunk boundary

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 - Detect and remove this as an edge case
- Some decoders will output impossible things like non-mod 4 4:1:0
 - Need to special case (different decoders do different things) and pad/crop as appropriate

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- Try to detect interlaced or telecined content based on frame content
 - Try to detect temporally distinct fields by how different even/odd lines are
 - Running sum spatially for field detection
 - Running sums temporally for previous, current, and future for added field order detection
 - Pattern detection for telecined content
 - e.g. CCNNC (reset at RAPs) for 3:2 pulldown

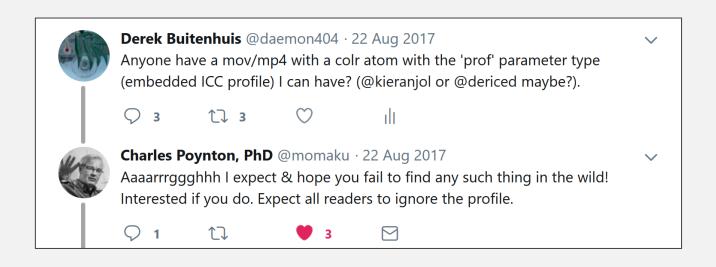
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- Try and detect "fake" 50i and handle appropriately

- More cringe warnings!
- Almost every non-HDR device requires BT.709 or SMPTE170M matrix/transfer/primaries
 - Try and convert based off tags if possible, try and fudge it otherwise (devices need color info)
 - Conversions need to be gamma-correct (swscale is terrible, use zimg¹ instead)
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- 10-bit input (not necessarily HDR!) is dithered accordingly with a random dither algorithm
- HDR input is ingested no matter what it is, although we only output HDR10-style media
 - Would have liked to use HLG, but Apple and Dolby exist
 - Need to make SDR versions, of course
 - Nominal peak luminance detection (zimg plays nice here)
 - In-house tonemapping like libplacebo's Möbius transform-based algorithm²

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- ICC profiles may be shipped with some codecs or containers
 - Extracted from e.g. MJPEG frames
 - ISOBMFF colr boxes may contain ICC profiles... in theory



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- We always deliver CFR content to clients/devices (come at me, bro), so need to choose a "good" rate
 - Quicktime bug-friendly, precision reduced to fit in e.g. H.264 or container fields
- Our indexing from earlier provides us with all the timestamp info for easy analysis
 - Analysis on DTS/PTS and frame durations (all done in arbitrary precision rational arithmetic)
 - Take into account allowed positive and negative timesamp discontinuities
 - Smooth extreme outliers (usually corruption) based on some metrics, e.g. stddev
 - If the file is an amalgamation of multiple CFR segments, choose the "best" rate
 - If it's "true" VFR, try and fudge a "good" middle ground rate

- Virtual timelines need to be taken into account
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- Make sure to take into account all things that can modify rates (e.g. mdia rate, trak rate, stts, edts)

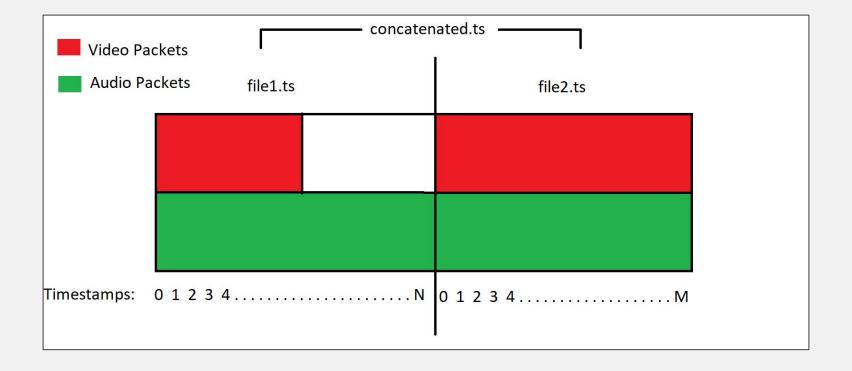
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- Resample when needed to a known-good set of sampling rates

- Cannot look at audio in isolation
 - Need the video to interpret audio timestamps correctly



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- Detect common mistakes like pasting captions into Word and upload...



Derek Buitenhuis @daemon404 · Oct 3

Today: Implementing .RTF detection in our captions stack. Because users.

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 - Spherical video, equirectangular video, 3D, ambisonics, etc.
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13

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- Working around plain old bugs

25 October 2018

Old Quicktime's H.264 decoder being unable to handle QPs less than 5



13

Links / References

- [1] zimg: https://github.com/sekrit-twc/zimg
- [2] Info on libplacebo's Möbius algorithm:
 - https://github.com/mpv-player/mpv/commit/d8a3b10f45eb10fb34ce9da3a9a76e3bd8644e3d
 - https://vimeo.com/album/5461208/video/293434018
- [3] Demuxed 2017 Talk: https://www.youtube.com/watch?v=cRSO3RtUOOk

Questions? Heckling? "Not a question, but..."?

15