P2PRG Live Streaming Research Questions

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Overview

- The upstream problem formulation
- Grid Topology constraints
- Push or Pull
- Static cost and Dynamic cost
- PET/IDA/Network Coding (the 'depth' question)?

RayV in a nutshell

Business:

> White label turn-key solution platform/CDN in a box for the Content owners/Telcos

> All content is legitimate

Working with Telco/MSOs/ISPs

Customers: NBA, DirecTV, Blizzard/Activision, Fox sports, American cap, Tennis Channel, Comcast CSN, AB Groupe, ex-pat channels, SMG

Usage:

> On average 500,000 connected peers

> 100,000 concurrent viewers at peak

➢ 8M minutes watched daily

P2P facts:

>90%-95% cost saving (HW and bandwidth)

> Building the network cost per concurrent viewer \$0.6

Monthly per concurrent \$0.5/month (10% of 1Mbps BW cost)

Improves quality (due to stream localization)

Customers requirements - Quality

Requirements:

- 500-800Kbps for news/music channels; 800-1.5Mbps for sports; 1.5Mbps to 3Mbps for TVHD experience.
- Flash cloud issues
- Adaptive quality (multiple qualities)
- Rules indicating who can contribute to who.

Requirement implication:

- Since peers can only contribute on average 200Kbps upstream P2P from peers only is limited to 20% of BW needed.
- Additional sources are needed
- These additional sources must contribute much more than they consume thus cannot be 'typical viewers' if consuming the entire stream.
- If those sources are 'CDN nodes' (hosted by an external CDN as in many 'hybrid solutions' or by the P2P provider) the P2P benefits are limited to 20%-30% only.
- Possible but 'not desired' solution: 'free riding' on high upstream peers such as universities and institutions.
- RayV solution: Adding many additional 'typical' nodes streaming to each of them minimal data (2 MTUs/sec) and having those re-distribute to many other peers (we call those 'amplifiers')

Upstream available



Quality and the upstream problem

Requirement: Streaming in at least 1Mbps.

Fact: Average upload from typical peer is 200Kbps.

1) Required BW = $N_V \cdot MR$

2) Available BW=
$$\sum UR_i = N \cdot \overline{UR}$$

3) Main goal function to maximize: $P2P_{Static} = 1 - \frac{U_S}{N_V \cdot MR}$ where U_S is upload from 'servers'

4) A crucial parameter describing the system is the difference between available bandwidth and

required bandwidth namely: $\beta = 1 - \frac{N \cdot \overline{UR}}{N_V \cdot MR}$

 β < 0 means that theoretically peers can support themselves.

If $\beta > 0$, β is the minimum needed to upload from servers regardless the method used.

5) Clearly if
$$N_V = N$$
 we get that $\beta = 1 - \frac{\overline{UR}}{MR} \approx 0.7 > 0$

6) Since the server needs to upload the stream at least once, we can define α as the amount of bandwith for the initial dispersion of the content to peers.

$$\alpha \ge \frac{MR}{N_V MR} = \frac{1}{N_V}$$

7) To summarize:
$$P2P_{Static} = 1 - \frac{U_S}{N_V \cdot MR} = 1 - (\alpha + \max(\beta, 0))$$

Customers requirements - Quality



Grid structure – Mash vs. Tree



- 1) Viewer must receive from each branch
- 2) Branch failure problem
- 3) Tree dynamic creation
- 4) Tree management, raising levels
- 5) Hidden assumption of 'balanced tree' or 'how to balance tree' and how to solve the 'weaker branch' problem?
- 6) Video layers complex things but solvable



- 1) Not efficient , how much?
- 2) Assuming PET/IDA/NC:
 - a) Depth problem
 - b) Delay due to segment size
 - c) CPU
 - d) SVC Layers complexity (solvable?)

Network coding and delay

$$A_{1} = \alpha_{1,1} \otimes P_{1} \oplus \cdots \oplus \alpha_{SZ,1} \otimes P_{SZ}$$
$$A_{2} = \alpha_{1,2} \otimes P_{1} \oplus \cdots \oplus \alpha_{SZ,2} \otimes P_{SZ}$$
$$\vdots$$
$$A_{223} = \alpha_{1,223} \otimes P_{1} \oplus \cdots \oplus \alpha_{SZ,223} \otimes P_{SZ}$$

P_i - is a media packet sized MTU Sz – Segment size **Considerations:** Segment size adds delay and CPU

Each coefficient needs to be sent in the protocol So new size is MTU+2^8*SZ*bits

<u>Delay</u>

1) Delay from encoder to viewers must not exceed 8-10 seconds

2) Viewers should watch 'simultaneously' with up to 2 sec difference

If a node receives a data chunk of size X every Y seconds, and spends the

Y seconds until the next chunk arrives uploading it to its acceptors than :

$$Y \approx D_{\text{Scheduling}}$$

There is a maximum total buffer between broadcast and viewing in live streaming which implies a maximum on $D_{Scheduling}$ and thus on Y :

$$(\mathbf{D}_{\text{scheduling}} + D_{Hop}) \cdot N_{hops} \le D_{Buff \, e\underline{r} \, \max}$$

Moreover, since we want the reliable sources as well for at least 2RTT + const we get :

$$(D_{\text{scheduling}} + D_{Hop}) \cdot N_{hops} + 2 \cdot RTT + \text{const} \le D_{Buff ermax}$$

Delay and hops

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Peer Dynamics

What is the cost of the dynamic behavior? – Peer churn

Ratio A/V = 3:1, 1000 Viewers average Viewer life time ~10 minutes No NAT traversal No delay in peer list



Network coding techniques

Sz - Segment size

Parameters⁻

y - number of primary amps, which receive their equations directly from the satellite.

w - 'mixing' factor. Each amp takes equations from w donors and generates a new on. naturally w, increases the needed upload from each amp.



Network coding techniques

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