

Adaptive Video Streaming: a Survey and Case Study

HU, Pili
MobiTeC, IE Department, CUHK
hupili@ie.cuhk.edu.hk
<http://personal.ie.cuhk.edu.hk/~hpl011/>

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Abstract

In the past decade, Internet traffic has seen a significant change from web browsing to video viewing. The ongoing trend raises a challenging problem: how to stream data to heterogeneous peers?

The designer of such data streaming architecture should bear the following considerations in mind: QoE, server load, network resource efficiency, scalability, etc. The heterogeneous peer network condition makes the design more complicated. The underlying codec ranges from Multi Description Coding to Multi Layer Coding. The data deliver architecture ranges from unicast, multicast, to P2P network. Researchers have focused on different system settings and optimization objectives.

This paper will first sum up several works in the context of adaptive video streaming. At the same time, we do a case study on a commercial adaptive video streaming system, which combines Multilayer Codec and P2P technology. Possible improvements on this system are proposed with reasoning. Some of the conjectures are verified through a corresponding simulation platform based on NS2.

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1 Introduction

- User perceived experience.
- Vendor cost.
- User cost.
- System cost.

2 General Model

3 Problem Scope

3.1 Codec

3.2 Networking

4 Design of Adaptive P2P VoD System

4.1 Codec Choice

4.2 Transimission Protocol Choice

4.3 Overlay Construction

4.3.1 Construction with Prior Knowledge

4.4 Peer Selection

4.5 Buffer Management

4.6 Chunck Selection

MMKP, Knapsack

MCMF, Network Flow

4.7 Playback Decision

4.8 User Model

5 A Case Study and Simulation

Our simulation platform roots from a commercial deployment of P2P system[40], which has upgraded from P2P file downloading system, to P2P live streaming sytem, and then to the current version of multilayer P2P VoD system. Interested readers are recommended to [13] for further information. This platform has been validated against real trace data collected from 2008 Olympic period. It is widely used in P2P related research and pre-deployment testing.

Course IERG5270[41] has adopted this platform as a simulation foundation for several years.

5.1 System Description

To get a concrete view of the situation, we draw the version tree in fig(1). Blue blocks represent the real deployment of ASTRI. White blocks represent corresponding simulation platform. Red dash stands for strong equivalency. Green block shows the position of this project. The trial version of multilayer P2P VoD is forked from the original downloader platform and is developed by Zheng Wen in 2010.

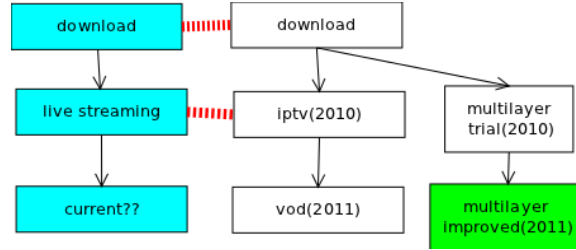


Figure 1: Version Tree of Real and Simulation

Precaution: the results obtained in this project may not reflect real system in every aspect(there is no red dash showing this kind of relationship). Nevertheless, methodology wise speaking, those results do provide some insights in Multilayer P2P system design.

5.2 System Benchmark Test

Before we start, system benchmark test of simulator performance is conducted.

Fig(2) shows how simulator running time varies with different number of cores. Note that PDNS [43] is the distributed version of NS[42]. In this paper, we just run the simulation on one machine with 16 cores. The original configuration of the platform involves 8 cores per simulation instance. Leaving some system recourses for other purpose, we can only run one simulation at a time, which significantly lowers our experiment efficiency.

We find that, when simulation time(measured in NS seconds) is small, 4 cores appear to be more efficient than 8 cores. Thus the following experiments are all issued to 4 cores to enhance our distributing level.

Fig(3) shows how simulator running time varies with the number of nodes simulated. It appears to be linear. The fit parameters are given in the title of the plot.

Fig(4) shows how simulator running time varies with the simulation time(in NS seconds).

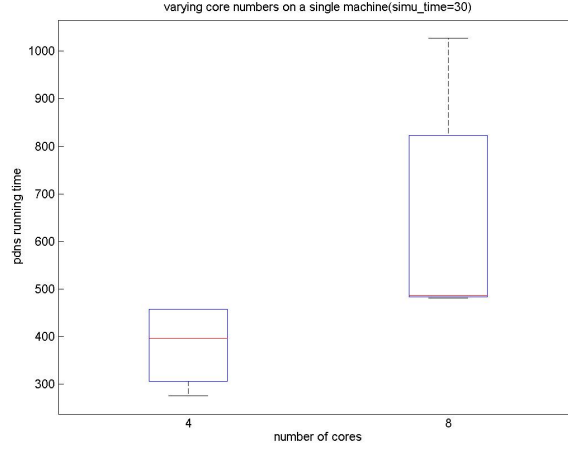


Figure 2: Running Time v.s. Number of Cores

With those system test, the final parameters we choose are:

- Number of Nodes: 160
- Simulation Time: 300 (NS seconds)

It's obvious that the number of nodes and simulation time is much smaller than those in a real VoD system. With those two configurations, the simulator can finish in about 2 hours using 4 cores. We sacrifice some equivalency to the real system to make the simulation tractable.

5.3 Experiment Configuration

In this study, we follow the same configuration given by Zheng Wen. The topology is summarized as follows:

- Star like. Peers are equally divided into 4 subnets. Every node connects to a single central router.
- Subnet parameters:
 - Subnet1: down:10Mb; up:10Mb.
 - Subnet1: down:3Mb; up:0.5Mb.
 - Subnet1: down:3Mb; up:0.5Mb.
 - Subnet1: down:3Mb; up:0.5Mb.

The layer configuration is summarized as follows:

- 3 Layers.

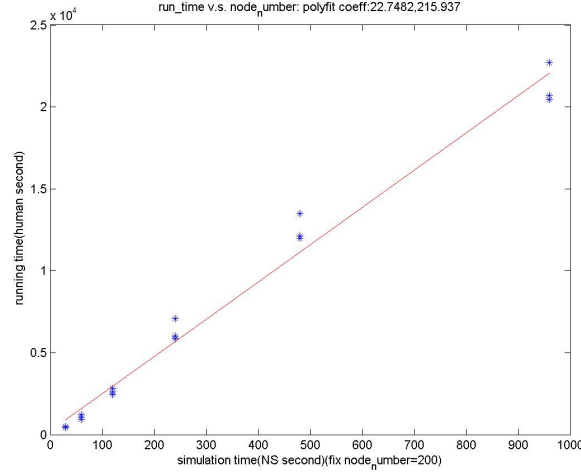


Figure 3: Running Time v.s. Number of Nodes

- Layer parameters:
 - Layer1: 256Kbit. (per piece)
 - Layer2: 256Kbit. (per piece)
 - Layer3: 512Kbit. (per piece)

Note that the cumulative data amount is 256Kbit, 512Kbit, and 1024Kbit. This is consistent with Wang’s QoE study[30]. Thus we can rely on their QoE model to measure our system.

5.4 QoE Model Implementation

In Wang’s paper[30], they seek for the tradeoff between bitrate(influenced by current layers subscribed) and discontinuity. They first do large amount subjective tests, and then fit the data into a continuous version formula, as is given in eqn(1).

$$MOS = c_1 \times d + \alpha \times (1 - e^{-b \times \lambda}) + c_2 \quad (1)$$

The parameters are $c_1 = -5$, $c_2 = 2$, $\alpha = 4$, $\lambda = 0.0015$.

In that paper, the maximum discontinuity considered is 50%, so we plot the MOS curve again to give the readers a better impression of how it looks like. See fig(5).

Fig(5(a)) is the 3D version and Fig(5(b)) is the contour version. We want to point out that by definition, MOS should be a score between 0 and 5. However, using this formula, we’ll get negative scores sometime. Especially when the discontinuity is 100% and bitrate is 0, the MOS can be as low as

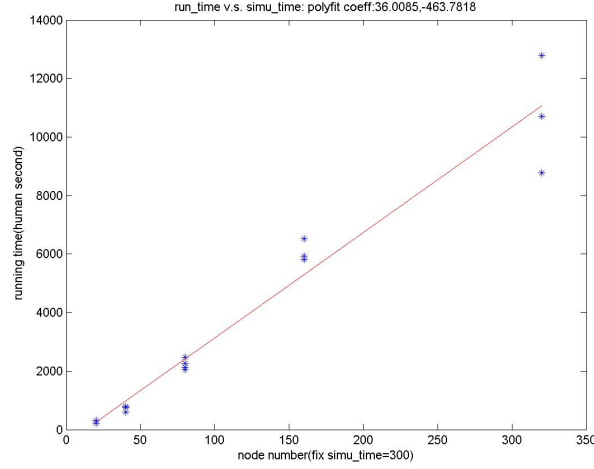


Figure 4: Running Time v.s. Simulation Time(NS)

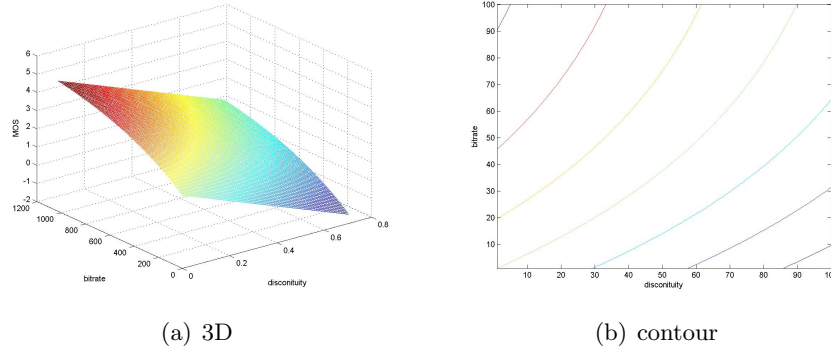


Figure 5: Conclusion, QoE and Performance

-3. We don't regard this case as a bug. On the contrary, it is reasonable to assign negative opinions. In our implementation of this QoE model, peers report their statistics every 20 seconds. If the -3 score happens, that means the user is purely waiting in the whole 20 seconds, which causes a negative opinion naturally. Our system should work towards reducing this kind of disturbance.

5.5 Baseline Test

In this section, we evaluate the baseline using the QoE records reported by peers. We compute MOS using eqn(1), and take the average.

Fig(6) shows the result when total simulation time varies from 30 NS seconds to 960 NS seconds. It's interesting that with the increase of simulation time, the QoE gets improved, and the boxplot shows the statistics of

5 simulations with the same configuration. This means the improvement is statistically significant.

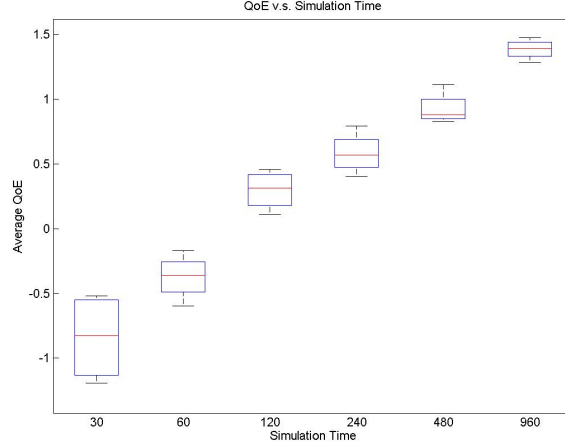


Figure 6: QoE v.s. Simulation Time

By examining the detailed trace file, we find that there are many negative opinions at the beginning of those simulation. Later after about 100 NS seconds, the reported MOS gets better and better. It makes sense that the system needs some bootstrapping time, when all of the peers lack chunks. They can not help each other efficiently during this time.

We eliminate those negative records and make the boxplot again. See Fig(7). This figure verifies our conjecture. The last two boxes stand for 480s and 960s simulation time. There is little difference in mean within this range. Longer simulation time just makes the variance smaller. To explain, after some bootstrapping period, the system approaches steady state and the average opinions become nearly a constant. The first box seems an outlier. This is because peers are joining the system gradually at that time. The number of reported records are very small. Only those peers connected directly to the source can get something to playback. Others all get negative opinions. By eliminating the negative opinions, the resultant score is very high.

Next, we pick a sample run out of all these baseline simulations. The global parameters and statistics are: simulation time is 480s; number of nodes is 160; number of QoE reports is 3749; number of nonnegative QoE reports is 2601; average QoE socre is 0.87; the nonnegative version of average QoE score is 2.02.

Fig(8) shows how QoE varies with curret time section. We can see the per section averaged QoE gets higher and higher in the first 150 seconds. After that, the section averaged QoE becomes steady. This again proves our

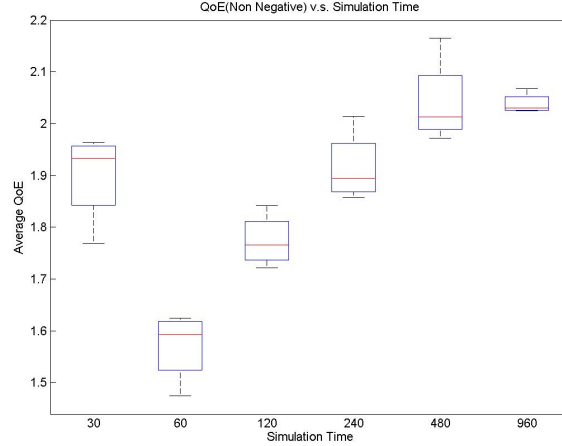


Figure 7: QoE(nonneg) v.s. Simulation Time

bootstrapping conjecture.

Fig(9) shows the histogram of QoE reports from this sample run. We can conclude that, some powerful peers can get full 5 score experience, whereas large amount of peers report negative scores. Our future improvements should mitigate this part(purely buffer for 20 seconds and play nothing).

5.6 Chunk Selection Architecture Reconstruction

In this section, we modify the original architecture to make it more clear and unified. This helps us develop strategies on this architecture later.

Original design separates chunk selection and peer selection. It is shown in Fig(10).

In this design, every peer is considered in a round robin fashion. After one neighbour is chosen, this peer selects chunks according to a three section based strategy. After one section is filled up to a certain ratio, the peer will consider the next section. Chunks in the same section are selected randomly. In our multilayer situation, layers are considered from lower ones to higher ones. That is, after the peer get 50% of the 3rd section, it will start to fill the next higher layer. In order to reduce the network burden, peers are constrained to fetch chunks from no more than 20 neighbours simultaneously.

The improved architecture is illustrated in Fig(11). To make our decision more rationale and make the architecture clear, we collect all information first, and do peer selection and chunk selection together.

In this version, we adopt a very simple strategy, as is used in PALS [26]. This strategy makes decision on a chunk basis. It first consider base layer chunks and then moves to enhancement layers. In one layer, the chunks

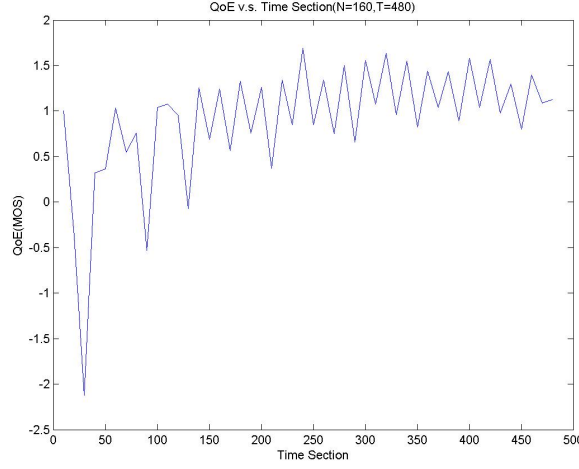


Figure 8: QoE Varies with Current System Time

closer to playback points are considered first. The overall picture is like a snake-shaped move on the 2 dimension buffer map. Fig(12) illustrates this strategy. If there are more than one peers available for one chunk, we just randomly choose one peer.

After each modification, we evaluate the result from two aspects:

- QoE. The formula is given in eqn(1). In our study, we didn't make any aggressive strategies like scheduling all requests to the server. So we don't subtract server's contribution from this metric intentionally. Another reason is, in the underlying overlay construction(which is not modified in this study), the server is treated as a normal peer. Not all peers can turn to server for help when the lack chunks.
- Performance. Another important thing we care is the performance of simulator. This kind of packet level simulation is very time consuming. One iteration of strategies calls for several hours of running typically. In order to fulfil a short-term project, we can not make strategies of high time complexity. What's more, the running time of simulator reflects the computation resource consumption in real deployments. User experience will degrade if their player consumes extra large amount of resources, to limit other process.

The result of modification in this section is:

- QoE: $0.7 \rightarrow 3.3$
- Performance: nearly doubled.

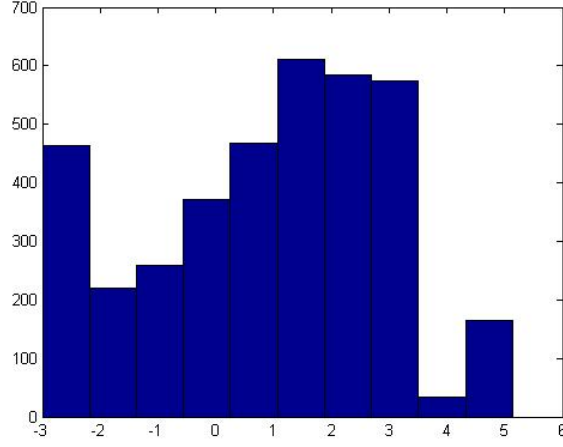


Figure 9: Histogram of QoE Distribution

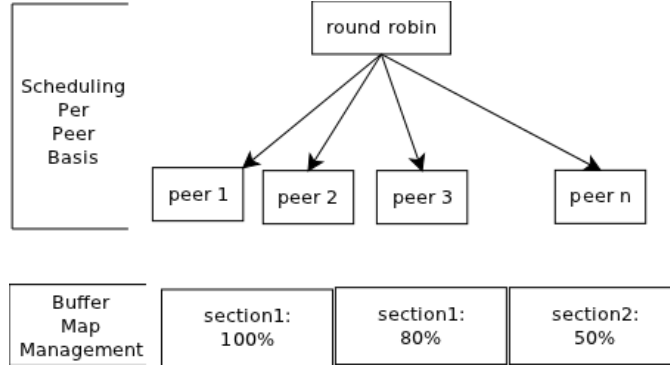


Figure 10: Original Architecture

5.7 Priority Based Upgrade

In this section, we do more engineering upgrades. We abstract the chunk selection strategy using a priority table. One sample table with window size equal to 5 is given in Table(1). This table is equivalent to PALS.

Further investigation on optimal priority table can be done easily in this framework. Result of this modification is:

- QoE: nearly the same.
- Performance: decrease slightly.

5.8 Scalable Window Size

After previous upgrades, we examine the trace files from several simulations. From the trace, we find many powerful peers can get full 3 layers in the

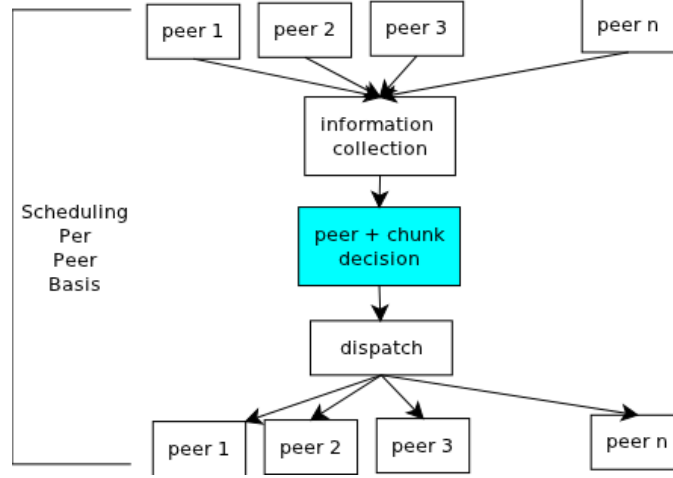


Figure 11: Improved Architecture

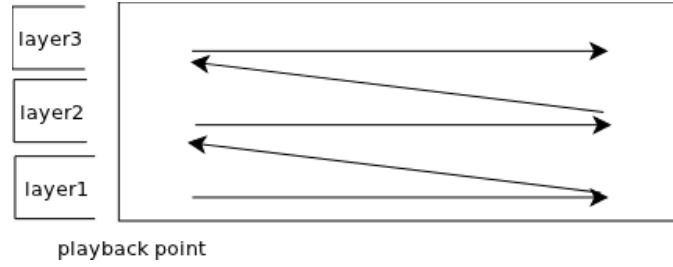


Figure 12: PALS Style Chunk Selection

whole window(60 seconds, 3 layers). The restriction on window size is a special desgin of video streaming, no matter live or VoD. This is because chunks very far from playback point is not of high interest to the peer. Every peer only needs to store enough chunks in the buffer to guarantee smooth playback. However, there is a side effect. Capable peers who are playing the early part can not help others who are playing the later part. Thus we want to give those peers a chance to help others. This is the rationale behind scalable window size.

Table(2) illustrates a scaled window with the size doubled. Using this table, less capable peers will act as they are using table(1). For those pow-

Table 1: A Sample Priority Table with Window Size = 5

	t1	t2	t3	t4	t5
layer3	11	12	13	14	15
layer2	6	7	8	9	10
layer1	1	2	3	4	5

Table 2: A Sample Priority Table with Window Size = 5+5

	t1	t2	t3	t4	t5	t6-t10
layer3	11	12	13	14	15	...
layer2	6	7	8	9	10	...
layer1	1	2	3	4	5	...

erful peers, after they finish the first section, as is in table(2), they can move to the second section. This is the simplest illustration of the idea of scalable window size. Ideally, the window size should be computed on recipient of heartbeat message. When a peer updates the network connection measurements, it can compute a proper window size based on those data. Last but not least, the priority table should be recomputed if the window size changes.

In this version, we fill the second section of the window using the same strategy as used for the first section. The result is:

- QoE is improved a little.
- Performance degrades sharply.

5.9 Performance Optimization

After last experiment, one simulation needs about 4.5 hours to complete. This is really time consuming. In order to do more iterations, we try to optimize the simulator performance right away. At first, optimization described in this section only aims at the performance of simulator. However, it turns out the QoE is also improved. We'll talk about this phenomenon in conclusion section.

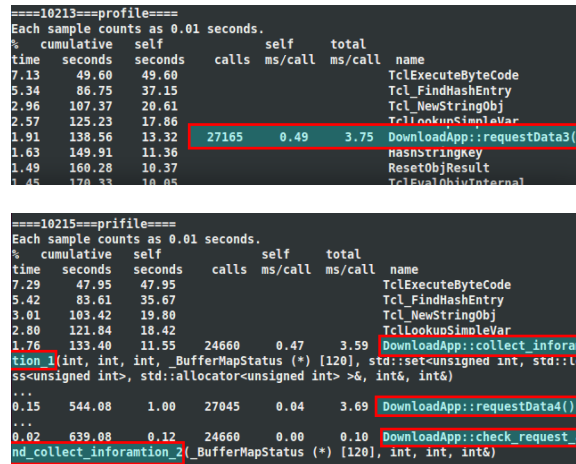


Figure 13: GNU Profile, Output

Fig(13) shows two screenshots of the output of 'gprof'. We first locate the most time consuming part that we can control. It turns out to be the 'requestData3()' function, which implements the strategy mentioned in last section. We then separate some sub functions from this one and find that the information collection is most time consuming. Note that we do not send probes to collect information. Instead, peer information is updated on recipient of periodical notification message. The information collection sub function only iterates over some map structures and fill the availability information into one unified 2D buffer map.

According to the analysis of the profile, we optimized those iteration process. For more details, please refer to the code repository of this project. Difference can be found between the following two commitments:

```
b45038404d8970c8ee1654d9bdbb10703432b8cd
5d3aabbdb1546d88da6e7a026d2d1fe67b04ad2f1
```

Result of the optimization is:

- QoE: improved a little.
- Performance: 4.5h \rightarrow 3.5h.

Note that this is not the end of the performance optimization. One of the possible directions is to optimize the sampling method using a reservoir[29]. However, this optimization is conducted with the knowledge that we'll only random select one peer if there are multiple peers available for one chunk. This kind of optimization is too aggressive and will limit the design space of strategies, so we don't do them in this research version. Designers can make notes and do this kind of optimization when all strategies are fixed in a real release version.

5.10 Introduce Randomness in Second Window Section

Due to time issue, our last trial based on the already established mechanism is to introduce randomness in second window section. As is known, randomness is a good property in the sense that it helps the system to spread chunks uniformly. The first window section is scheduled using a snake-shaped priority, this design considers user-perceived experience. However, the second window section is for those powerful peers. Those chunks are not hurry for playback. Thus we make the second section of the priority table randomly initialized. Every peer will have different priority table.

The result is:

- QoE: get worse.
- Performance: get worse.

5.11 Conclusion of the Case Study

In this section we put together the result of 6 big versions. See Fig(14). In terms of QoE, version 2 improves a lot from baseline. After that, there are not so much difference in QoE. In terms of simulator performance, all modified versions are slower. The slowest version is the scalable window one. From the plot, we know our performance optimization really helps to reduce running time.

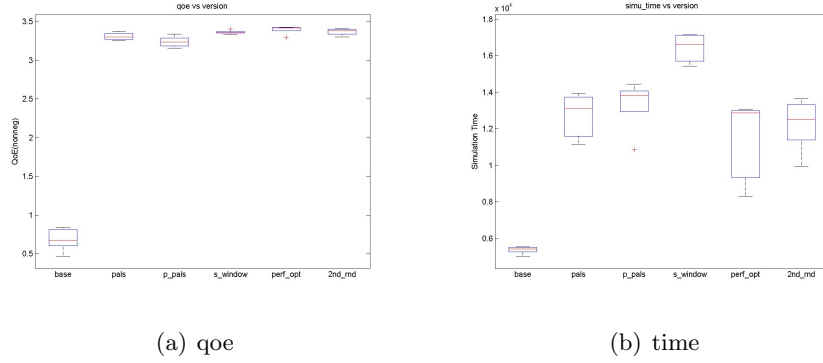


Figure 14: Conclusion, QoE and Performance

Fig(15) gives a closer look at the 5 modified versions. The most interesting observation is the difference before and after performance optimization. Note that in section(5.9), we did nothing about strategy. It is simply equivalent code reconstruction. However, while we observe better performance of the simulator, we also find the QoE is improved a little. From the box plot, this improvement seems statistically significant (there are two outliers).

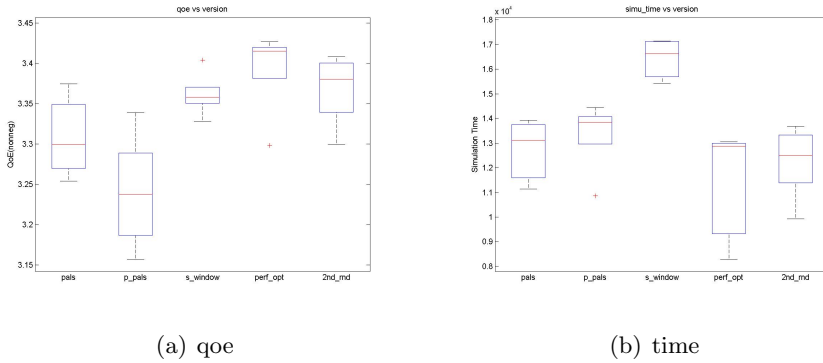


Figure 15: Conclusion, QoE and Performance

Similar phenomena have been observed in our other VoD simulations using the same series ASTRI platform[13]. When the server is idle, the results tends to be better. Just like in our experiment, after performance

optimization, the computation burden is lowered, thus we see better results. One conjecture is that this phenomenon is due to underlying timing facilities, either NS[42, 43] layer or the p2p simulator layer[13]. The verification of this point is left as future work.

6 Conclusion

- Engineering approach v.s. academic approach:
where is the biggest cake?
- Time distribution:
 - 70%, literature survey.(30+ papers)
 - 15%, bugfix of the platform, environment setup.
 - 5%, first unified version(QoE:0.7→3.3, biggest improvement in this study)
 - 10%, scalable window, performance optimization, random 2nd section. (little outcome)
- 6 big versions / 240 runs.
- Auxiliary Scripts:
 - .sh:298 lines
 - .pl:791 lines
 - .m:133 lines
- Simulation Code Difference:
 - download_agent.cc: 1940 lines
 - download_agent.h: 301 lines
 - labtesting.tcl: 84 lines

7 Future Works

Here we list some potential research topics in an adaptive video streaming system:

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For multilayer P2P VoD system, we list some potential research topics:

- Composition and decomposition degree.

- If we use the framework given in section(5.7), what's the optimal priority table. In order to achieve QoE aware design, models like [30] should be considered. Due to time limit, we haven't deducted the optimal table in this study.
- As for the simulator, how to explain the relationship between QoE and simulator performance. Theoretically speaking, QoE should only depend on strategies and irrelevant of simulator performance.

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Appendix

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