A Experiment over Media Player Application based on Dash.js

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Abstract

A video streaming programm called DASH offers on-demand video playback. It has a web server that runs on HTTP. The streamed video is divided into equal-length chunks. These segment objects are obtained by the client each time an HTTP request for a clip is made. Every video has an own Media Presentation Description (MPD) file, which the client uses to learn about the portions that make up a specific video.In this experiment, we employ a media player approach that relies on dash.js (v4.5.0), which transitions actively and plays media streamlets while playing a stored DASH playlist or the mpd file that we want. The effectiveness of the built-in Adaptive BitRate (ABR) algorithms is then tested and compared using a variety of experiments. Some of the algorithms provided are throughput-based, dynamic, and BOLA.By keeping track of key parameters like bit rate, buffer level, throughput, and segment properties like length and time, the efficacy is calculated.

1 INTRODUCTION

The experiment focuses on the three ABR algorithms such as throughput-based, dynamic and Bola algorithm. We have used the testplayer.html page in the dash player to conduct this experiment. The three algorithms are checked in their corresponding html files. All in all we have three testplayer.html files defined and one of the three algorithms is enabled in one of the html files. We then record the network metrics for all the three algorithms by switching the network conditions between Fast3G and no throttling.

1.1 ABR Algorithms

ABR gives a quicker starting time along with smoother feeds. The video can start playing right away because ABR streams often start with a low bitrate stream before the selection algorithm determines the available bandwidth or buffering space. As HTTP is used for delivery, most web servers and CDNs are compatible with HTTP-based ABR streaming technologies. It is therefore more affordable than putting up customized services or maintaining constant connectivity. In Throughput-based ABR, streaming related decision is performed based on the throughput details from the previous downloads of the application.BOLA ABR algorithm uses the details of buffer level at the client side for decision making.While downloading a segment, BOLA utilises a bitrate se-

lection mechanism that converts the buffer level—the overall amount of seconds of video segments saved in the buffer—to the bitrate of the following segment. Dynamic ABR algorithms uses both the factors mentioned above.

1.2 Performance Metrices

1.2.1 Selected BitRate

Bit rate is the measure of how much information is sent through a communication connection in a specific length of time. The term "bit rate" is frequently used to characterise the transmission quality of music and video in the context of streaming. While requiring more bandwidth and storage space, higher bit rates typically provide audio and video of greater quality. In conclusion, bit rate is a term used to define the quantity of data delivered per unit of time and the quality of streams. Here we are calculating bitrate in Mbps.

1.2.2 Buffer Level

A specific amount of data is put into the buffer when a video file is started before playback starts. More data is stored into the buffer as playback continues in order to keep the buffer level constant. The video can delay or stutter when the player tries to load additional data into the buffer if the buffer level starts to drop. The quantity of video data that is temporarily kept on a user's device during a video streaming session before being seen on the screen is referred to as buffer level. In situations where the network connection may not be quick enough or reliable enough to provide the video data in real-time, this buffer is utilised to make sure that the video playing continues smoothly and without interruption.

1.2.3 Measured Throughput

The pace at which data is delivered across a network connection while a video is being played back is referred to as measured throughput in video streaming. It is a measurement of the real network transmission capacity for streaming video, accounting for network performance-altering variables such as congestion. The player may occasionally send queries to the streaming server for brief bursts of information (referred to as segments) in order to track throughput during video streaming. The quantity of information received divided by the amount of time required is used to compute throughput. Next, the duration required for each segment to download is noted.

1.2.4 Segment Download Time

The length of the period required for a chunk of video data to download from of the streaming server to the user's browser is referred to as the segment download time in streaming video. A segment is a brief section of the video file that is downloaded independently from the remainder of the movie. With segmented video downloads, the video player can continue playing the clip well before complete file has finished downloading.

1.2.5 Segment Size

In streaming video, the term "segment size" describes how much data is contained in each segment. The quality and bit rate of the clip, the speed of the network connection, and the capacities of the client browser are only a few of the variables that might affect the segment size. Larger segments often contribute to speedier playing since receiving and analyzing each segment take less time. But, lengthier download periods brought on by bigger chunks may also increase the chance of stuttering or other problems during playing.

2 METHODOLOGY

2.1 Metrics Collection

In the main.js file of the testplayer under low latency, we modified the METRIC_INTERVAL attribute from 1000 to 8000 milliseconds so that the setInterval method iterated every 8 seconds. The BitRate and Buffer is retrieved using the metrics defined the current testplayer page elements, Figure 1. The metrics are returned in html tags and in order to get the required value, we extract the inner value of the tag. The

```
//Bit Rate
console.log('Bitrate:', +self.domElements.metrics.videoBitrate.innerHTML);
bitRateArray.push('Bitrate: ' + self.domElements.metrics.videoBitrate.innerHTML + '\r\n');
//Buffer
console.log('Current Buffer:',self.domElements.metrics.bufferTag.innerHTML);
bufferArray.push('Current Buffer: ' + self.domElements.metrics.bufferTag.innerHTML + '\r\n');
```

Figure 1: Bit Rate and Buffer retrieval.

measured throughput is a result returned by getAverageThroughput method defined under the player object held by the current test player object, Figure 2. The segment attributes are retrieved using the HTTP object, Figure 3. To fetch the current HTTP object we have utilised, getCurrentHttpRequest method defined under dashMetrices.

```
//Throughput
let throughput = self.player.getAverageThroughput('video');
console.log('Measured throughput:',throughput);
throughputeArray.push('Measured throughput: ' + throughput + '\r\n');
```

Figure 2: Measured Throughput retrieval.

The absolute difference between finish time and the response time would give us the total segment time. The responseHeader of HTTP object inculcates various information including the content length too. This content length is nothing but the segment size that we are interested in. The metrics that we read is stored in respective array. After the entire video is played, we put the results into single array and then stored as BLOB into a text file, Figure 5. We have

```
//Download Time
const httpObject = dashMetrics.getCurrentHttpRequest('video');
const segmentTime = Math.abs(httpObject._tfainish.getTime() - httpObject.tresponse.getTime());
console.log('Segment download time: ' + segmentTime + 'ms');
downloadArray.push('Segment download time: ' + segmentTime + 'ms' + '\r\n');
//segment size
console.log('Segment size',httpObject._responseHeaders.split('\r\n')[1])
segmentSizeArray.push('Segment size: ' + httpObject._responseHeaders.split('\r\n')[1] + '\r\n');
```

Figure 3: Segment time and length calculation.

```
Performance Metrics:
selected bitrate (Mbps), buffer level (second), measured throughput (Mbps), segment download time (second), and seg
            ncy testplayer with
                                               Low latency testplayer with
  Dynamic Algo
                                              Bola Algo
                                                                                         Throughput Algo
  Example showing how to use dash.js to
                                                                                         Example showing how to use dash.js to
  play low latency streams. All low latency
                                              play low latency streams. All low latency
                                                                                         play low latency streams. All low late
  related settings such as the different
ABR algorithms(LoL+,L2A) are
                                               elated settings such as the different
                                                                                          related settings such as the different
                                              ABR algorithms(LoL+,L2A) are
                                                                                         ABR algorithms(LoL+,L2A) are
```

Figure 4: Three HTML files for three ABR Algorithms.

maintained three HTML files for the three ABR algorithms, Throughput based, dynamic and BOLA. This is achieved by enabling the radio button value specific to the algorithm to checked as showed in Figure 6.

```
var blob = new Blob(result, {type: 'text/plain;charset=utf-8'});
saveAs(blob, 'metric output result.txt');
clearInterval(interval);
}
```

Figure 5: Storing the metrics to file.

Figure 6: Setting Bola radio button to checked.

The metrics are recorded with changing network conditions. The first 3 minutes of video is played with 3G, then next minutes by switching to No throttling, next 2 minutes with 3G again and rest of the playtime with No throttling.

3 RESULTS AND CONCLUSION

On no Throttling network condition, the throughput based ABR algorithm, might choose a higher video bitrate to offer a better viewing experience. On fast 3G network, the algorithm reduces the video bitrate to fit within the available network capacity. There might be chances of not having a good video quality but on the positive side, it avoids buffering and continuous streaming of the video. The Dynamic algorithm can

choose a higher video bitrate in a no-throttling network to improve the viewer experience with a buffer level of at least one second. The Dynamic method may need to reduce the video bitrate in a Fast 3G network in order to maintain a buffer level of at least one second and match the available network capacity. Although the video quality can suffer as a result, buffering won't occur and playback will be seamless. Using Fast 3G, Bola can adjust the video bitrate to the available bandwidth utilization while maintaining a constant buffer level. If the network efficiency increases, Bola may then raise the video bitrate. The Bola technique simply chooses a higher bitrate on a no-throttling network to improve the watching experience while keeping a constant buffer. For the optimum streaming experience for the user, a total of three algorithms modify the video bitrate depending on the available network capacity. The Throughput method, which is the most basic algorithm, performs well with Fast 3G and no network throttling. The Bola method enhances the video quality while keeping a steady buffer level, while the Dynamic method takes into mind the buffer threshold to prevent buffering.

As a part of future work, we can improvise the way of selecting network conditions. Instead of having three html files, we can include a function that can change the network condition for a specified playtime of the video.

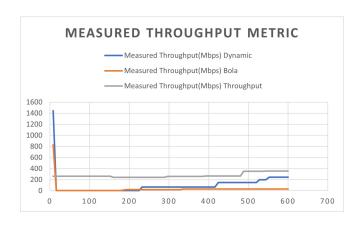


Figure 7:Measured Throughput recordings in Mbps.

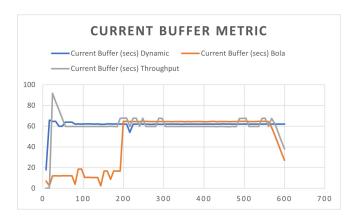


Figure 8: Buffer Metric recordings in secs.

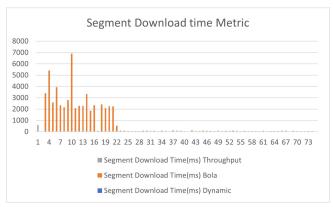


Figure 9: Segment download Time recordings in ms.

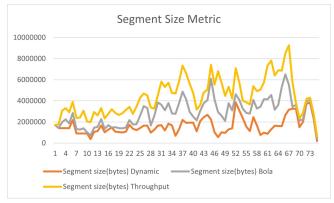


Figure 10: Segment size recordings in bytes.

References

- [1] GitHub link of this experiment, https://github.com/dinesh-07/dash.js/tree/development/samples/lowlatency/testplayer,
- [2] Abdelhak Bentaleb, Bayan Taani, Ali C. Begen, Christian Timmerer and Roger Zimmermann, A Survey on Bitrate Adaptation Schemes for Streaming Media Over HTTP, https://ieeexplore.ieee.org/stamp/stamp.jsp?arnumber=8424813
- [3]Saamer Akhshabi, Ali C. Begen and Constantine Dovrolis, An Experimental Evaluation of Rate-Adaptation Algorithms in Adaptive Streaming over HTTP, https://www.cc.gatech.edu/fac/Constantinos.Dovrolis/Papers/final-saamer-mmsys11.pdf