RESEARCH ON MODELING REAL-TIME MULTIMEDIA STREAMING USING HLS PROTOCOL

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ABSTRACT

With the technology innovation in electronic devices, the rising demand for high-quality multimedia content will continually grow. Over the last years, there is a shift from the traditional approach of streaming the media content to the modern approach of adaptive bit-rate streaming, which offers the benefit of switching to the best quality video according to varying network conditions or available bandwidth thus providing the enriched user experience. The adaptive streaming methods are more or less wholly based on HTTP, so can efficiently work over large distributed networks. This paper aims to deploy HLS (HTTP Live Streaming) for streaming of real-time multimedia, to support mobile as well as desktop browsers without the use of any plugin and also to deliver multimedia with reduced latency, therefore allowing rich quality media content to be delivered over a large number of CDN (content delivery networks). Few added advantages of using HLS are: it can easily traverse through any firewall or proxy server, easier to cache, authentication and encryption (encryption is optional which can be done using AES-128), and recoverability from an outage.

Keywords: - *HLS, TS, CDN, Streaming, Transcoding, AWS, S3, EC2*

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1. INTRODUCTION

HLS is an adaptive bitrate technology. Adaptive streaming is referred as multiple variants of source video encoded at different bit-rates. When a video is encoded to HLS, multiple files are generated for varying bandwidths and different resolutions. The generated streams are mapped to the client in real-time using .m3u8 master playlist file (also referred as index file) according to the screen size to be supported and available bandwidth.

HLS allows sending audio-video content over HTTP from an ordinary web server for client playback. One can send live streams or video on demand, the source can be supplied through USB webcam, RTSP Source, or H.264 files. In the case of live broadcasts, CDN that is being used understands that the .m3u8 index files are not to be cached for more than one media segment interval (index file changes dynamically). HLS provides many

alternate streams at different bit-rates, and the client software can switch streams cleverly as network bandwidth changes. Browser plays HTTP Live streams natively as the source for the <video> tag.

1.1 How HLS Works:

To model multimedia content using HLS, no special streaming server is essential, hence all the switching logic be present in the player. A software encoder inputs media source, encodes it into supported video and audio codecs, and yields MPEG-2 transport stream, which is then directed to the stream segmenter to output a series of short media segments (say segment_1, segment_2 … segment_n; refer Fig-1). All the generated segments are linked with the Index file.

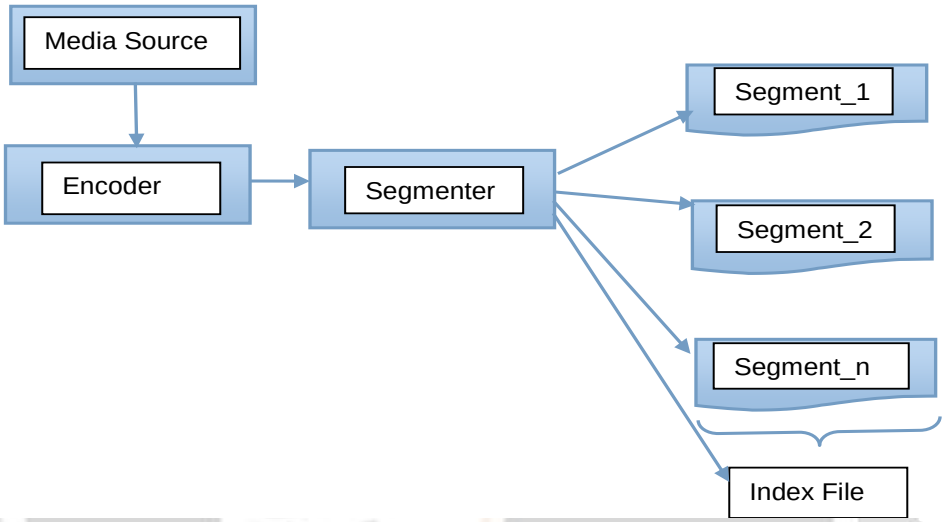

Fig-1: HLS Working

Session types:

You Can Send Live Streams or Video on Demand, the source can be supplied through USB camera, RTSP, or H.264 files.

HLS Server Content Preparation:

A product encoder (say ffmpeg) takes sound video input, encodes it as H.264 video and AAC sound, and yields it in a MPEG-2 Transport Stream, which is then sent to programming stream segmenter to deliver a progression of short media sections and spared as a progression of .ts documents.

HLS Distribution:

The segmenter likewise produces an ace playlist (i.e. file record) which contains a rundown of the media documents. The URL of the file document is distributed on the HTTP web/reserve server, which can read and transmit data according to HTTP responses. Client requests the listed Index files sequentially.

2. LITERATURE SURVEY:

TABLE-1: Literature Survey Comparison Table [1, 2, 3, 4, 5]

Sr. No.	Paper Title	Method / Tools	Limitation / Future Plan
1	Dynamic HTTP Live Streaming Method for Live Feeds	Dynamic transcoding algorithm for HLS	Formation of Adaptive Master document in light of past solicitations by different customers
2	Implementation of Continuous HTTP Live Streaming using Playback Position Request Mechanism in heterogeneous Networks	Web-Based Multimedia Player using pure web standard technology	Multimedia services on smartphone with multiple NICs
3	Optimizing the Resource updating Period Behavior of HTTP Cache Servers for Better Scalability of Live HTTP Streaming Systems	Nginx cache server	The optimized solution is proposed only for Nginx server
4	A new mobile streaming protocol based on HTTP live streaming protocol	Open source application FFMPEG	The difficulty of implementing time-shifted system is the organization of index file.
5	Implementation of HTTP Live Streaming for an IP Camera using an Open Source Multimedia Converter	Utilized an open source library named FFMPEG	Latency of 30 seconds

3. PROPOSED SYSTEM:

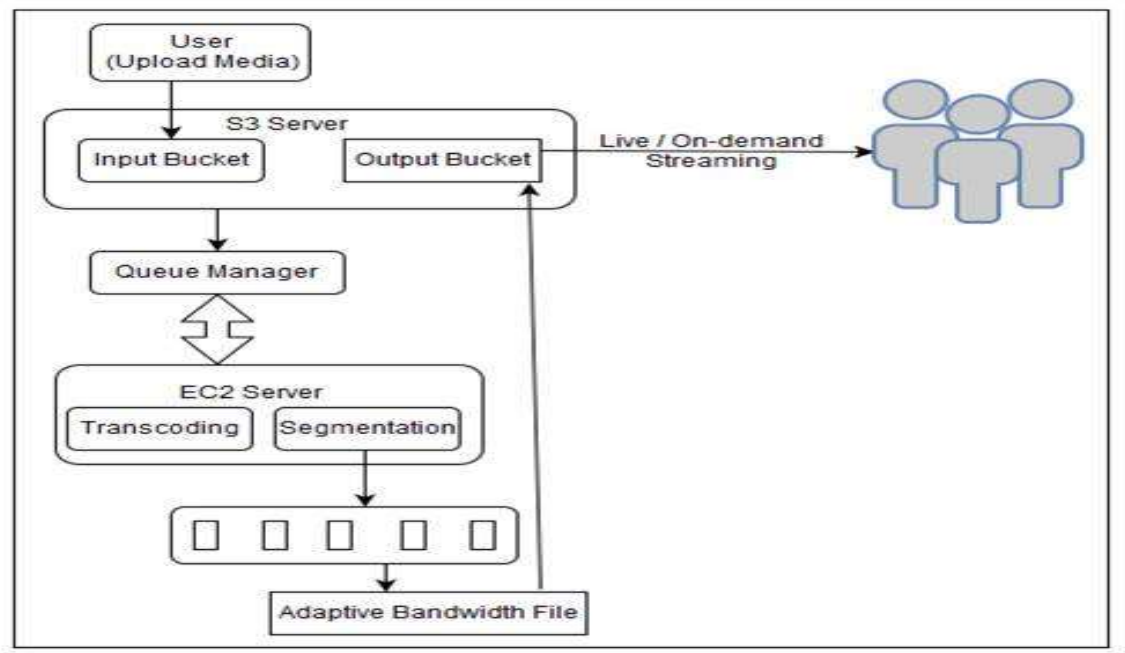

Fig-2: Proposed Architecture

S3: Amazon's (S3) Simple Storage Service offers boundless online storage space for documents or data of any type. One can use S3 to securely store the private files, or as an information storage component in dispersed network application architecture. S3 also offers access control mechanisms that permits to keep the information private or make it open and available to everyone on the internet.

Bucket: An S3 bucket is a container to collect data objects. A bucket is an appropriate mode of grouping objects collected (here objects store data or metadata).

EC2: Amazon's (EC2) Elastic Compute Cloud gives surroundings to run as many or as limited virtual servers as requirement arises.

Work-Flow:

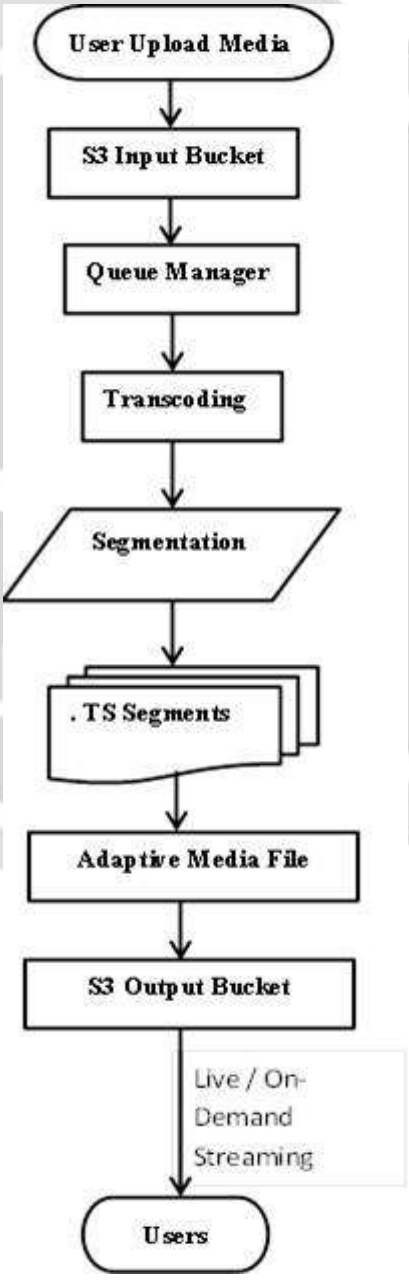

Fig-3: Flow Chart

4. IMPLEMENTATION ENVIRONMENT AND RESULTS:

1. Eclipse (Java Language)
2. Sample Video
3. FFMPEG(Video Converter Tool)
4. Knowledge of AWS services(S3 and EC2)
5. Supported media player to playback m3u8 stream (E.g. JW Player, VLC)

FFMPEG can be used for following purposes:

- It can read-write Audio/Video file formats and decompress-compress their content.
- It can effortlessly parse a file to any format, convert it to a different format and even transmit it through a network via any protocol.
- It can parse the fragmented MOV file transcoding packages into a MPEG-TS container.
- It also supports multiple outputs created out of same input in the same process.
- E.g. A sample video can be transcoded in 240P, 360P, 480P and more using FFMPEG.
- FFMPEG Mediafilesegmenter tool - Creates TS segments from MP4 files.

4.1 Output Snapshots:

Output 1:

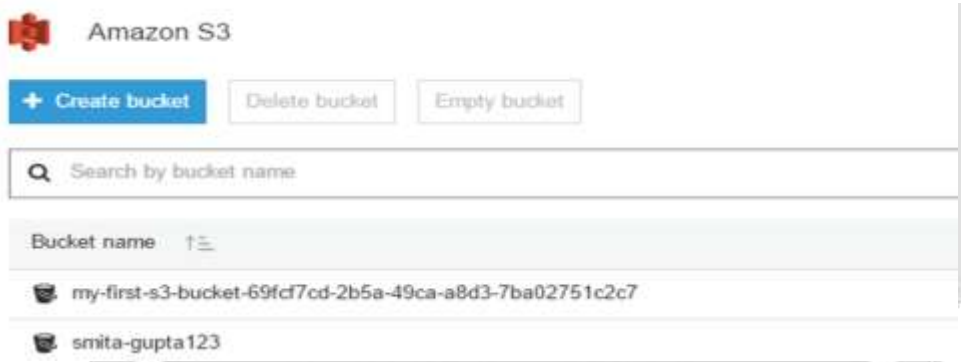

Fig- 4: Bucket Creation

Output 2:

Fig-5: Data Segmentation

Output 3:

Fig-6: Generated Mater Index File

Output 4:

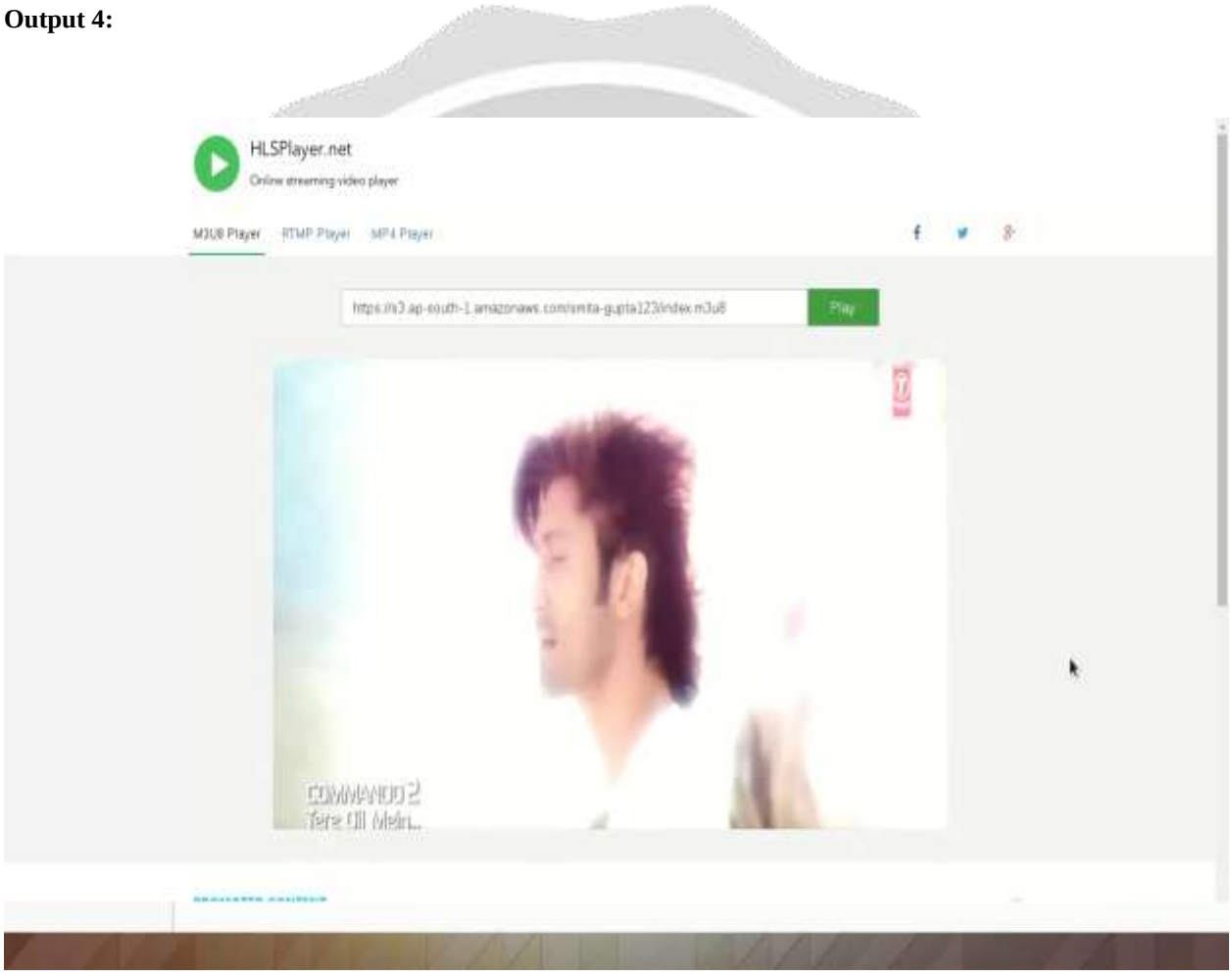

Fig-7: On demand video streaming

Output 5:

Fig-8: User Login

Output 6:

Fig-9: Call Establishing

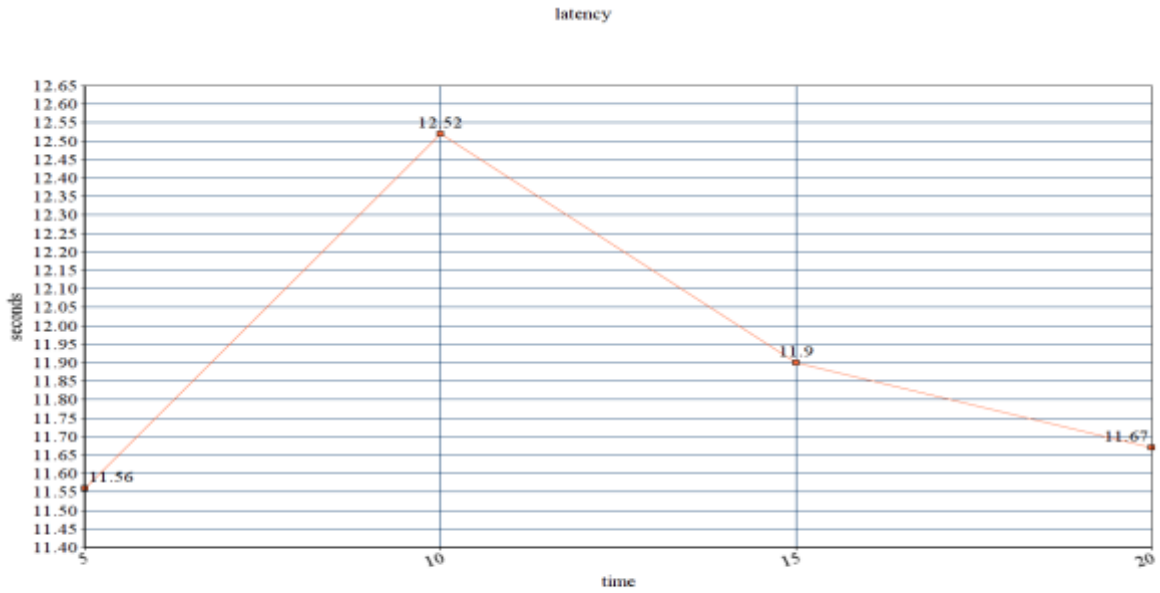

Fig-11: Graph representing latency at different time breaks

Fig-11 represents latency in seconds at time gaps of 5 minutes (x-axis) for four unlike values. On the base of these values, the middling latency that we get is 11.9 seconds.

5. CONCLUSION

In the video streaming applications whether it is on-demand or live video streaming there exists various number of evaluation parameters, amongst them the utmost essential ones are latency and the video quality. In this dissertation work an attempt is made to deploy http live streaming for streaming of multimedia content with the objective to provide media with reduced latency for real-time video streaming set-ups (e.g. video conferencing). Thus the proposed scheme can be used to deploy HLS for streaming of real-time as well as on-demand multimedia to be viewed by users with the reduced latency.

6. ACKNOWLEDGEMENT

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