

# A Novel Approach to QoS Provision in a Cloud-Based Multimedia Storage System by using An Automated Resource Allocation

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## ABSTRACT:

Cloud computing is a relatively new trend in Information Technology, which is growing rapidly, that involves the provision of services, which we call them as resources, over a network such as the Internet. Mobile Computing is another area where mobile devices such as smartphones and tablets are believed to replace personal computers by combining network connectivity, mobility, and software functionality. In near future these devices will take over the traditional desktop and laptop devices and effortlessly switch between different network providers. To maintain network connectivity all the time different service handover mechanisms may be used so that user can access cloud services without interruption. But if user mobility is considered then as user is moving geographically to various locations with these mobile devices then he is connected to its local cloud to access the cloud services. Because of this mobility factor, network congestion may increase which causes degradation of QoS. And different Cloud providers are not in a position where they can easily build multiple

Clouds to service different geographical areas like they do with services that run on individual servers. Hence, a new method for

service delivery will take into account which will improve QoS in order to provide better QoE to the clients and better load management to the providers, as well as helps to reduce network congestion on a global scale. This paper gives a insight that as the demand for specific services increases in a location, it might be more efficient to move those services closer to that location Using an analytical framework. This will help to reduce high traffic loads due to multimedia streams and will offer service providers an automated resource allocation and management mechanism for their services.

**KEYWORDS:** Mobile computing, Web services, Communication system traffic control

## 1. INTRODUCTION

Cloud computing is a computing paradigm, where a large pool of systems are connected

in private or public networks to provide dynamically scalable infrastructure for application, data and file storage. With the advent of this technology, the cost of computation, application hosting, content storage and delivery is reduced significantly. Security issues in cloud concerns mainly associated with security issue faced by cloud service providers and the service issues faced by customers. In cloud computing, users can outsource their computation and storage to servers (also called clouds) using internet. This frees users from the hassles of maintaining resources on-site. Clouds can provide several types of services like applications (e.g. Google Apps, Microsoft online), infrastructures (e.g., Amazon's EC2, Eucalyptus, Nimbus) and platforms to help developers write applications (e.g., Amazon's S3, Windows Azure). Much of the data stored in clouds is highly sensitive, for example, medical records and social networks. Security and privacy are thus very important issues in cloud computing. In one hand, the user should authenticate itself before initiating any transaction and on the other hand, it must be ensured that the cloud does not tamper with the data that is outsourced. When clients access multimedia objects from a content server, the content server must have sufficient disk and network to deliver the objects to the clients. Otherwise, it rejects the requests from the new clients. Thus, the popular content server can easily become the bottleneck in delivering multimedia objects. Therefore, server and network workloads are important concerns in designing multimedia storage systems over the Internet. Multimedia objects, like other traditional data files and Web pages, may be transferred across networks, such as the Internet. In order to provide efficient delivery of data across the networks, some data can be stored in the

middle of the network. When requests for the same object have been received, the sedate can be used to satisfy the requests at the middle of the network instead of forwarding the request any further. This method to satisfy requests with previously accessed data is called caching. Since caching needs to consume a certain amount of storage space, the cache performance is affected by the size of the cache memory. Cloud computing refers to the delivery of computing resources over the Internet in which large groups of remote servers are networked to allow the centralized data storage, and online access to computer services or resources. Cloud services are made available to users on demand via internet from cloud provider's machines which we call them as servers. These servers are designed to provide easy, scalable access to applications, services and resources and are totally managed by cloud service providers. A cloud service provider offers some component of cloud computing typically Infrastructure as a Service (IaaS), Software as a Service (SaaS) or Platform as a Service (PaaS).

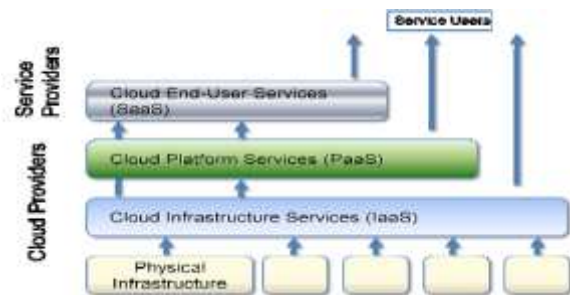


Fig1: Cloud Infrastructure

Mobile devices (e.g., smart phone, tablet pcs, etc) are increasingly becoming an essential part of human life as the most effective and convenient communication tools not bounded by time and place and can access services through wireless network.

Mobile Cloud Computing is a promising solution to bridge the widening gap between the mobile multimedia demand and the capability of mobile devices. Thus, when it comes for data available in the form of videos Cloud allows its customer not only to access videos that are on demand but also application in the form of services to view and manipulate. This paper is based on Cloud Based Mobile Media Service delivery. In this, services are mainly populated on local clouds. As per the user's demand idea services on local cloud have the capability of moving these services to the nearby clouds. This may result in less traffic congestion on network which results in improving QoS on the network and also providing better QoE to the clients.

## II. RELATED WORK

Media-Edge Cloud (MEC) is recently proposed architecture which improves the performance of cloud technologies. This architecture aims to improve the QoS and Quality of Experience (QoE) for multimedia applications. This is achieved by a "Cloudlet" of servers running at the edge of a bigger Cloud. This architecture mainly handles requests closer to the edge of the Cloud and hence helps to reduce latency. If further processing is needed, then requests are sent to the inner Cloud, so the "Cloudlets" are reserved for QoS sensitive multimedia applications. This aims to divide the network hierarchy within the Cloud, in such a way that physical machines that are closer to the Cloud's outer boundaries will handle QoS sensitive services. Since these machines reside on the border of the Cloud, the data has to travel less distance within the Cloud before it is sent out to the clients. This not only improves QoE for clients but it also reduces network congestion within the

Cloud. But, this concept of MEC does not take into account user mobility, for improving cloud performance. Furthermore, all the research at present assumes that only one entity (the provider) is in control of a Cloud and as a result different providers cannot "share" resources in a manner that can improve the utilization efficiency of their hardware. This can potentially lead to problems in the future as mobility and multimedia-rich content becomes more popular and high bandwidth data streams will have to travel great distances and reach moving targets. Cloud providers may find themselves in situations where their hardware resources are not adequate and they may have to create more Clouds to handle the load and relieve network congestion.

## III. SYSTEM ARCHITECTURE

If the storage space is large, more objects can be stored on the cache storage and the probability of finding an object in the cache is thus high. The cache performs better. If the storage space is limited, only a few objects can be stored in the cache storage, and the probability of finding an object in the cache is low. As a result, the cache performance becomes low. Therefore, the cache size influences the cache performance. Since caching stores some previously fetched objects on the storage devices, the presence of an object exists on the storage devices significantly affects the efficiency of the caching. When a new object is being accessed, the cached mission policy decides whether an accessed object should be stored on to the cache devices. Since the cache performance increases monotonically with the number of objects in the cache, the cache storage space is often full in order to keep the most number of objects in the cache. When an accessed object

needs to be stored and the cache space is full, the cache replacement policy decides which object should be deleted from the cache storage to release space.

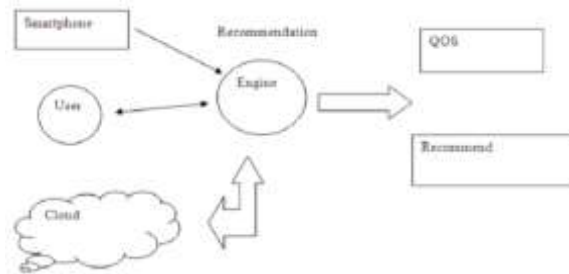


Fig 2: System Architecture

The majority of QoS research at the network layer has focused on the QoS routing. A multimedia application often has stringent requirements on the delay. QoS routing determines the delivery path for flows taking into account both the availability of network resources and the QoS requirements of the flows.

There are active researches in providing QoS aware routing algorithm for mobile multimedia applications. Researches in the recent QoS routing in mobile ad hoc networks have been covered. One of the major functions at the transport layer is congestion control and the TCP protocol is the dominant protocol at the transport layer. TCP protocol is designed for the wired networks and is not efficient for wireless networks. It reduces the transmission rate when there is a packet loss, which suffers great performance degradation since the wireless channel generates a higher bit error rate. The transport layer protocol should be able to differentiate the packet losses generated by the congestion and by the channel errors.

#### IV. IMPLEMENTATION

There are several factors that need to be considered in the provision of QoS, such as the limited resources on the mobile devices, heterogeneity, and roaming characteristics in the mobile computing environment. Power-efficient design of QoS is the common solution to address the limited battery on the mobile device. Context-aware middle-ware is used to overcome the heterogeneity issue in the mobile networks and provide context-aware QoS. Handover is essential in mobility management which provides QoS when the mobile devices move from one network to another network. The evolution of new applications and technologies, such as social multimedia and cloud computing, poses many challenges in the provision of QoS. In order to gather QoS data and know the network conditions in a specific area, we are using another mechanism that we call the QoS Monitor.

The power-aware multimedia solutions jointly design the video coding parameters and channel parameters to adapt to the video contents and underlying network conditions to minimize the total energy consumption. An efficient system should jointly consider three factors: bit rate, power consumption, and video quality. A balance needs to be achieved between power consumption in computation and communication to provide energy efficient multimedia applications. The goal is to minimize the total power consumption, subject to three constraints: the maximum video distortion to ensure satisfactory video quality, maximum end-to-end delay required by the application, and the maximum computational complexity provided by the mobile multimedia devices. Another goal is to minimize the video distortion, subject to the maximum power consumption allowed, maximum end-to-end delay, and maximum computation complexity. Power-rate-distortion analysis adds a new dimension

power to the traditional rate-distortion analysis.

The complexity parameters of the video encoding scheme can be dynamically adjusted to maximize the video quality under the energy constraint of the mobile device. It is considered to be part of the SCL and acquires such data by querying the clients for network conditions. The mechanism that we are assuming here that can resolve human-friendly service names to unique Service IDs. In the SDL we need mechanisms that will connect service subscribers to the correct instance of a service for service delivery purposes. A record of Service IDs and in which Clouds their instances are running and also uses input by the QoS Tracking are maintained by the Service Tracking and Resolution or STAR. STAR will make a decision on which Cloud is better suited to service a client request based on the location of the client, using this information. STAR achieve this functionality is by look up routing tables in order to identify which Cloud is closer to a user. Service to reject the new client and forward them to another Cloud if possible. This gives control to service providers and also becomes a contingency mechanism in case STAR makes a wrong decision. The STAR server can be scaled similarly to the DNS system since it is essentially the same type of service albeit with some extra parameters. Once a Cloud ID is found, then the ID is resolved into the IP addresses of the Cloud controllers that the client can contact to access the service. The process is shown in the Fig. 2. It should be noted that alternatively the Cloud ID can be returned to the client, at which point, the client will have a choice of which DNS to use to find the IP addresses.

## V ANALYTICAL APPROACH

For the first, we define the time to prefetch blocks of data, which is given by:

$$T_{\text{Prefetch}} = L + C * p \text{ eq. (1)}$$

In this equation, L is the network latency and C is the per block time of copying data between the in-cache memory and network buffers. Ideally should be at least equal to the number of blocks required to display a video frame of data. On a lightly loaded wired network we can consider these values constant for each link. However, in a mobile environment, changes as the client moves and the number of network links increase.

We can express L as follows:

$$L = F_{n,s,\theta} + F_{\text{cloud}} + F_{\text{Protocol}} \text{ eq. (2)}$$

Where,  $(F_{n,s,\theta})$  is the latency incurred by the number of links(n) between client and service, the network bandwidth on each link ( $S_i$ ) and the network load on each link ( $\theta_i$ ),  $F_{\text{cloud}}$  is the Cloud latency caused by the network topology and hierarchy within the Cloud  $F_{\text{protocol}}$  is the latency caused by the transport protocol. If the time to prefetch blocks is larger than the time it takes for the device to consume them, then we have jitter. This can be expressed as:

$$T_{\text{prefetch}}(p) \geq T_{\text{cpu}} * p \text{ eq. (3)}$$

Where,  $(T_{\text{cpu}})$  the time it takes for a device to consume a number of blocks by playing them as audio and video frames.  $(T_{\text{cpu}})$  is therefore dependent on the type of video being displayed and the hardware capabilities of the mobile device.

We now substitute for  $T_{\text{prefetch}}$  in (3) with the expressions in (1) and (2). Rearranging,

we get:

$$F_{n,s,\theta} + F_{\text{cloud}} + F_{\text{Protocol}} \geq (T_{\text{cpu}} - C) * p$$

Exploring network latency in detail, for each link we have transmission delay and queuing delay. Therefore, the total network latency will be the sum of the latencies for each link between client and service. Hence, we can express as:

$$F_{n,s,\theta} = \sum(D_{ti} + Q_i)$$

If we denote the transport block size as  $b$ , then the time to transmit  $p$  blocks over a link is equal to the number of blocks multiplied by the block size and divided by the bandwidth of the link.

$$F_{n,s,\theta} = \sum((p * b) / S_i + Q_i)$$

$$F_{cloud} + F_{Protocol} + \sum((p * b) / S_i + Q_i) \geq (T_{cpu} - C) * p$$

On a lightly loaded system, we consider  $F_{Protocol}$ ,  $F_{cloud}$  and  $Q_i$  to be negligible

$$\sum(b / S_i) \geq (T_{cpu} - C)$$

Let be the soft limit that we are aiming for in order to prevent jitter and SL is the migration time.

$$HL - SL = atMt$$

Where,  $a$  is the rate of network latency increase as the number of network links increases. We can calculate  $a$  at the mobile device and we can also find  $Mt$  between two Clouds.  $HL$  is given by the mobile device, so we can calculate to SL find where to set out QoS trigger for service migration. We can visualized how the increasing number of links between a user and a service can bring the connection near the QoS limit and how we can use a soft limit to trigger service migration in order to prevent this. We can also see that for a given migration time, we

need to adjust SL so that during the migration the QoS will not reach the HL[1].

## VI CONCLUSION

Efficient search on encrypted data is also an important concern in clouds. The clouds should not know the query but should be able to return the records that satisfy the query. Accountability of clouds is a very challenging task and involves technical issues and law enforcement. Neither clouds nor users should deny any operations performed or requested. In this paper, we studied the challenges which are faced by the mobile user in future networks. The service delivery models which are used currently are not that much sufficient and not consider the needs of mobile user in future. A cloud storage system was proposed in order to provide robust, scalable, highly available and load-balanced services. In the meantime, the system also needs to provide quality of service provision for multimedia applications and services. The proposed system achieves the QoS in distributed environment which make the proposed system especially suitable to the video on demand service. It often provides different service quality to users with various types of devices and network bandwidth. We believe that our implementation will provide the better quality of service (QoS) as well as better quality of experience (QoE) to the user.

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