

DiffServ RED Evaluation with QoS Management for 3G Internet Applications

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Abstract

With the increasing commercial deployment of wireless networks, the issue of providing multiple services is becoming more and more important. So the concept of "Quality of Service (QoS)" is being widely discussed and implemented. In the 3G architecture, the UMTS has defined 4 kinds of service types to provide the appropriate QoS type for different requirements and different applications. Similarly, in order to provide QoS in the traditional IP network, we assume 3G networks will adopt DiffServ as the IP-backbone.

In this paper, we proposed an efficient mapping from 3G services to DiffServ PHB aggregates. We used queueing theory to estimate the delay and loss for different traffic types. By using the estimation, we proposed a homogeneous QoS mapping policy to achieve QoS requirements under efficient resource utilizations through the 2 different services. An Admission Policy with QoS mapping is suggested to assure QoS by compromising with little throughput degradation. Besides, we also propose the adaptation policy. This adaptation policy could dynamically adapt the RED queue based on the arrival traffic types. Thus, we perform the QoS management through the RED evaluations. The QoS management includes the mapping and adaptation. For further development, we may also use these models to estimate and propose other mapping policies such as profit based mapping.

1. Introduction

Wireless communications and Internet have grown tremendously for the past decade, and will be converged for providing ubiquitous integrated services. The 2nd generation (2G) mobile communication system, which is widely operated now, is mainly used for transmitting voice. It is inefficient to support data service on Internet. In addition, the data rate is only 9.6Kbps, it is not sufficient to support some applications which require more bandwidth or throughput. So the 3G system[1,2] is developed. 3G wideband technology increases data rate, which is 384kbps in the normal or walk situation, 128kbps in the vehicle situation and 2Mbps in the fixed situation. In order to support real time services, e.g. voice, video, end-to-end QoS[3,4] is an important part of this evolution. In the specification of 3GPP (3rd Generation Partnership Project), the UMTS[5] has defined the QoS architecture and 4 different services which can support different QoS requirements for different applications.

Traffic	Example of App.	Capacity reservation type
Conversational	Voice, Video telephony	Static
Streaming	Real time streaming video	Static
Interactive	Web browsing, Real time control channel	Dynamic
Background	Down of files and mails	Dynamic

Table 1. UMTS Service Classes

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The main distinguishing factor between these QoS classes is how delay sensitive the traffic is: Conversational class is meant for the traffic which is very delay sensitive while Background is the most delay insensitive traffic class.

The QoS concepts are also taken into consideration in traditional IP network. IETF has two groups that are discussing the QoS on Internet, one is Integrated Service (IntServ)[6] and another is Differentiated Service (DiffServ)[7]. IntServ uses RSVP to reserve network resource to assure the service quality. And the DiffServ is between Best Effort and RSVP, which makes the network more scalable and resource usage more efficient.

In differentiated service architecture[8,9], classification and conditioning functions of traffic are implemented only at boundary nodes entering the DiffServ Domain (called Ingress nodes). The ingress node marks the TOS (Type of Service) of each packet according to policy service provisioning. After being marked at the boundary node, packets are forwarded by appropriated PHB[10,11] on each node within the DS domain. Both boundary and interior nodes must be able to apply the appropriate PHB to packets based on the DS codepoint (DSCP). DiffServ has defined 3 different PHBs that are EF, AF and Best-Effort to achieve the different QoS requirements.

In this paper, we assume DiffServ QoS model is adopted by IP backbone. Since the UMTS has defined 4 different service classes, all mobile applications will be marked as one of these classes. However mobile Internet sessions attempting to access to the DiffServ Domain will incur some problems. It is due to some gap between 3 DiffServ PHBs and 4 3G services. So, how to map the 3G services to DiffServ PHB aggregates will be a policy decision for the 3G operators. In this paper, we propose a mapping policy to achieve QoS requirements under efficient resource utilization through the 2 different domains.

The rest of the paper is organized as follows: Section 2 presents the QoS framework and architecture for our mapping

policy. In Section 3, we propose the traffic model for estimate the delay and loss. This estimation is used for admission policy and mapping decision. Section 4 discusses the mapping issue and presents the homogeneous mapping policy. Finally in Section 5 we present a summary and point out our future research work.

2.QoS and Mapping Framework

This section we propose the QoS framework and the mapping interface for our mapping policy. The network architecture is shown as Figure 1.

The GwR/BB is the gateway that is responsible for the call admission and resource allocation. Packets are marked as one of the UMTS service classes in UMTS Domain. These traffic classes provided by the 3G wireless networks should be mapped to the 3 DiffServ classes before they enter the DiffServ Domain. Thus, the GwR is the interface connecting to the DiffServ Internet backbone, where the SLA is negotiated to specify the resource allocation by the 3G operator to serve the aggregate traffic flowing into the gateway.

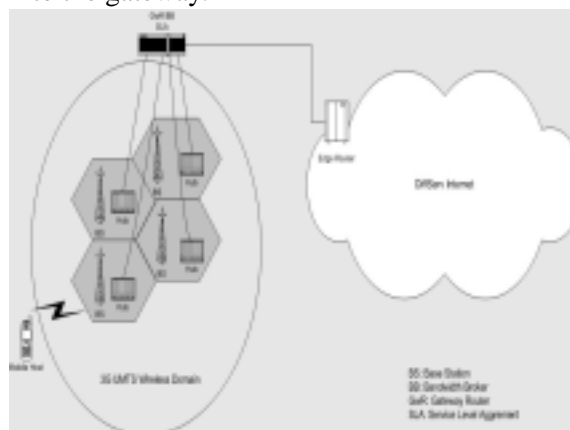


Figure 1. *QoS Architecture*

And here we propose our mapping interface shown as Figure 2. This Mapping Interface should exist in the GwR and make the mapping decision for the 3G operator and thus for the call admission and resource allocation.

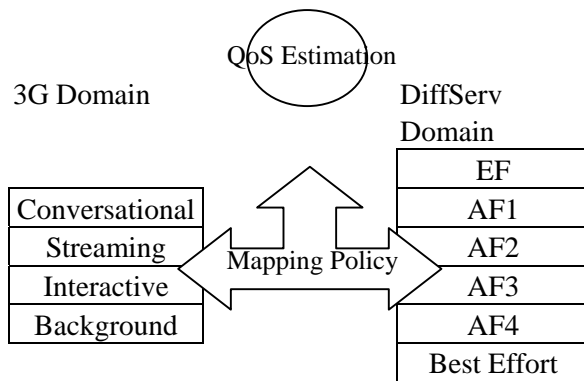


Figure 2. Mapping Interface

The goal of mapping policy is to achieve QoS requirements (data rate, delay, jitter and loss) for both existing and new sessions. In other words, a new 3G session should be mapped to a certain PHB aggregate with acceptable QoS for all sessions within such an aggregate. Thus, we need to estimate the delay and loss for admitting this new session and for choosing the PHB aggregate. In section 3 we introduce queueing delay model to calculate these. We also use 3 traffic models for 4 3G UMTS services. As shown in Fig 2.2, the Conversational and Streaming are mainly used on real-time applications that have higher delay requirements. Interactive and Background are used on traditional Internet applications such as WWW, E-mail or Telnet. From these attributes of each service, we model the Conversational and Streaming as CBR traffics; Interactive and Background as Poisson or Exponential On/Off traffics. So we model the queues as D/D/1, M/D/1 and Exponential On/Off models. In section 4, we propose the mapping policy for making the mapping decision. The mapping policy uses the queueing delay model to decide how to map a UMTS session to DiffServ Domain.

3. Traffic Model with Queueing Delay

Analysis

This Section we will analyze and evaluate the RED (Random Early Decision) queue [12,13] with different model. RED

queue is a kind of queue mechanism that the DiffServ suggest to implement the AF queue. These models can be used on the different traffic types, such as Poisson or CBR...etc. UMTS had defined 4 classes each belongs to different application types. Thus, our model can be used for the policy-maker to determine the delay and jitter of a given traffic.

We consider a router with a queue size K . With the RED queue management scheme, arrival packets are dropped with a probability that is an increasing drop function of the average queue size.

A typical drop function is defined by four RED parameters-- \min_{th} , \max_{th} , \max_p , and w , where the w is usually a fixed and small parameter in RED.

The average queue size is estimated using an exponential weighted moving average:

$$avg_k = (1 - w) avg_k + w k$$

then the typical drop function is :

$$\left\{ \begin{array}{l} drop(avg_k) = 0 \text{ if } avg_k < \min_{th}, \\ drop(avg_k) = 1 \text{ if } avg_k \geq \max_{th} \\ \text{otherwise,} \\ drop(avg_k) = \max_p \cdot \frac{avg_k - \min_{th}}{\max_{th} - \min_{th}} \end{array} \right.$$

3.1 RED with D/D/1/K model

While the packet arrival is CBR (Constant Bit Rate), then the queue model will become D/D/1/K. In this section we will discuss the delay and loss of the D/D/1/K queueing model.

3.1.1 queueing delay and loss estimation

Assume that the inter-arrival time is $1/\mu$, and the service time is $1/\mu$. Then we define the parameter n as :

$$n = \mu / \mu$$

The n means the numbers of arrival packets in each service time.

Here we take $n \geq 2$, because when the n bigger than 2, this system is thought

super-overloading, and the arrival packets will be dropped with probability almost 1. So we take $n = 2$ as a reasonable and normal load.

Then, by using Imbedded Markov Chain [14], we could get :

$$\begin{aligned}
 p_{a,a-1} &= [\text{drop}(a - 1)]^n \\
 p_{a,a} &= \sum_{i=1}^n \text{drop}(a)^{i-1} * (1-\text{drop}(a)) * \\
 &\quad \text{drop}(a+1)^{n-1} \\
 p_{a,a+1} &= \sum_{i=1}^n \sum_{j=i}^n \text{drop}(a)^{i-1} * (1-\text{drop}(a)) * \\
 &\quad \text{drop}(a + 1)^{j-i-1} * (1-\text{drop}(a+1)) * \text{drop}(a + 2)^{n-j}
 \end{aligned}$$

The $p_{a,b}$ is one-step transition probability, i.e. the probability that a departure packet sees “ b ” packets in system given that the previous departure packet sees “ a ” packets in system.

The transition probability matrix $\underline{\underline{P}}$ is as follows:

$$\underline{\underline{P}} = \begin{bmatrix} p_{00} & p_{01} & p_{02} & 0 & \dots & \dots & 0 \\ p_{10} & p_{11} & p_{12} & 0 & \dots & \dots & 0 \\ 0 & p_{21} & p_{22} & p_{23} & 0 & \dots & 0 \\ 0 & 0 & p_{32} & p_{33} & p_{34} & 0 \dots & 0 \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \\ \dots & \dots & \dots & \dots & \dots & \dots & \dots \end{bmatrix}_{(K+1) \times (K+1)}$$

then we can use this matrix to compute iteratively:

$$\underline{\underline{d}}^{(j+1)} = \underline{\underline{d}}^{(j)} \cdot \underline{\underline{P}} \quad (1)$$

where $\underline{\underline{d}}$ is probability vectors $[d_0, d_1, \dots, d_K]$, d_i is probability of seeing i packets in system when a packet departs the system.

In this paper, we use a C-program to compute the stationary probability vector $\underline{\underline{d}}$. After computing $\underline{\underline{d}}$, we can estimate the system delay. In D/D/1/K model, we can let $d_i = r_i$, where the r_i represent the probability of seeing i packets when a new arrival enters the system.

Then we can calculate the average system delay by:

$$\sum_{i=0}^K r_i \cdot \frac{1}{\mu} \cdot i + \text{mean residual time} \quad (2)$$

Here for simplicity we approximate the mean residual time as a half of service time, so the

$$\text{mean residual time is } \frac{1}{2\mu}.$$

Then we consider the loss probability of this model. The loss probability of the RED queue will be:

$$\text{loss_probability} = \begin{cases} \frac{\lambda - \mu}{\lambda} & , \text{when } \lambda > \mu \\ 0 & , \text{when } \lambda < \mu \end{cases} \quad (3)$$

3.1.2 Multiple Flows Estimation and

Compare with NS

Now we consider the case of multiple flows entering into the same D/D/1/K queue. For simplicity again, here we still view the aggregated arrival traffic as deterministic. That is, the load will be:

$$\rho = \frac{\sum_m \lambda_m}{\mu} \quad (4)$$

And we can use the above D/D/1/K RED model to estimate delay in the case of multiple flows.

Next we give an example and compare it with ns2 results.

Example: Suppose there are m CBR sources that enter the RED queue, the arrival rate are from λ_1 to λ_m and service rate is 5Mbps. The RED queue size is 40 with parameters $\text{min}_{th}=10$, $\text{max}_{th}=30$ and $\text{max}_p=0.1$. The queue size is the same as max_{th} .

The environment in NS2 is as follows:

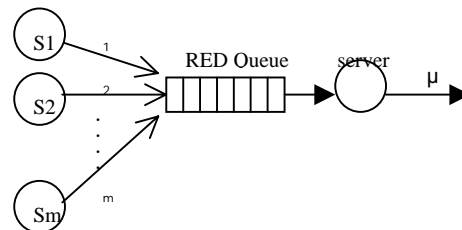


Figure 3. NS2 Simulation Environment

Now we make a contrast between our models and ns2 from 1 flow to 5 flows. The results are shown as follows:

No. of multiplexing flows		Delay	Loss	Difference
1	NS	0.02428	0.332	Delay < 1.7%
	D/D/1/K	0.02388	0.333	Loss < 0.3%
2	NS	0.02194	0.0904	Delay < 4%
	D/D/1/K	0.0214	0.090	Loss < 1%
3	NS	0.021239	0.087	Delay < 1 %
	D/D/1/K	0.02149	0.090	Loss < 4%
4	NS	0.02169	0.0897	Delay < 0.1%
	D/D/1/K	0.02149	0.090	Loss < 0.3%
5	NS	0.02187	0.0904	Delay < 1 %
	D/D/1/K	0.02149	0.090	Loss < 0.4 %

Table 2. Comparison with D/D/1/K and NS2

3.2 RED with M/D/1/K model

This section we discuss the M/D/1/K model where the arrival is Poisson and service rate is deterministic. Poisson is a general arrival type for many applications. And deterministic service rate can be used for the AF class of the DiffServ Domain. Because we use the Weighted Round Robin scheduler to serve the AF sub-classes (The RFC suggest we use 4 sub class for AF — AF1 to AF4), each service time of the queue of sub-class is deterministic.

3.2.1 Queueing Delay and loss estimation

Firstly, we have to compute one-step transition probability $p_{a,b}$, the calculate scenario is the same as D/D/1/K model.

The p of M/D/1/K is as follows:

$$P_{a,b} = \sum_{i=b-a+1}^K f(i; \frac{\lambda}{\mu}) \cdot C_{b-a+1}^i \cdot (1 - drop(a-1))^{b-a+1} \cdot drop(a-1)^{i-(b-a+1)}$$

$\forall a > 1$ and $b > a-1$, where $f(i; \frac{\lambda}{\mu})$ is a Poisson distribution

$$P_{a,b} = 0$$

$\forall a > 2$ and $b < a-1$

And the matrix \underline{P} is:

$$\underline{P} = \begin{bmatrix} p_{00} & p_{01} & p_{02} & p_{03} & \dots & p_{0K} \\ p_{10} & p_{11} & p_{12} & p_{13} & \dots & p_{1K} \\ 0 & p_{21} & p_{22} & p_{23} & \dots & p_{2K} \\ 0 & 0 & p_{32} & p_{33} & \dots & p_{3K} \\ \dots & \dots & \dots & \dots & \dots & \dots \\ 0 & 0 & 0 & 0 & \dots & p_{KK} \end{bmatrix}_{(K+1) \times (K+1)}$$

Again, we use equation (1) to compute \underline{d} , and equation (2), (3) to get the delay and loss.

3.2.2 Comparing with NS

Again, we make a contrast between the M/D/1/K model and NS2. The NS2 simulation environment is the same as D/D/1/K (Figure 3) besides the arrival type becoming Poisson.

Because arrival is Poisson, the multiple flows estimation can be easily modeled by simply summing up these Poisson arrivals. Then we still make the contrast between NS and the M/D/1/K model.

No. of multiplexing flows		Delay	Loss	Difference
1	NS	0.0239386	0.36	Delay < 0.1%
	M/D/1/K	0.023943	0.37	Loss < 2%
2	NS	0.030063	0.424	Delay < 0.5%
	M/D/1/K	0.03023	0.428	Loss < 1%
3	NS	0.0234149	0.32	Delay < 2%
	M/D/1/K	0.023943	0.37	Loss < 15%

Table 3. Comparison with M/D/1/K and NS2

3.3 RED with Exponential ON/OFF

Traffic

This section we will discuss the traffic type—Exponential ON/OFF. This traffic type generates traffic according to an Exponential ON/OFF distribution. Packets are sent at a fixed rate during ON period, and no packets are sent during OFF period. Both ON and OFF periods are taken from an exponential distribution and packets are constant size.

3.3.1 Queueing Delay Estimation

Because packets are sent at a fixed rate during ON period, we can take it as CBR traffic. Thus, we separate this case into 2 parts and analyze the 2 parts respectively. First part is the ON-period, we could use the D/D/1/K model to estimate the delay due to the characteristic of Exponential ON/OFF traffic. Second part is the OFF-period, we know that no packet is sent during OFF period, but, we have to consider the tail effect when system load is bigger than 1. For simplification, we assume that the buffer is exactly full when entering OFF period, and the average delay is the mean buffer size multiply the service time. This is the **idle time compensation** that compensates the delay caused by the remaining packets when entering OFF period. Suppose the mean time of ON period is x ms, the mean time of OFF period is y ms and arrival rate is λ during ON period, we can estimate the delay as:

$$\frac{x}{x+y} \times avg_ON_delay + idle_time_compensation$$

where the avg_ON_delay is estimated by D/D/1/K model with arrival rate λ and $idle_time_compensation$ will be:

$$= \begin{cases} 0 & ,if\ load < 1 \\ meanbuffer\ size \times service\ time & ,if\ load > 1 \end{cases} \quad (5)$$

3.3.2 Multi-Flow Estimation

At the beginning, we analyze the case of multiple flows from two. So we suppose there are 2 exponential on/off traffic where the mean time of ON period is x_1 and x_2 , the mean time of OFF period is y_1 and y_2 . The arrival rate during ON period is λ_1 and λ_2 . Then we firstly consider the probability that both exponential on/off traffics are during On period. Given source 1 is during On period, the probability of seeing source 2 during On period is:

$$\frac{\lambda_1}{\lambda_1 + \lambda_2} \times \frac{x_2}{x_2 + y_2}$$

Similarly, given source 2 is during On period, the probability of seeing source 1 during On period is :

$$\frac{\lambda_2}{\lambda_1 + \lambda_2} \times \frac{x_1}{x_1 + y_1}$$

So the probability that both flows are during On period is as follows:

$$p_{both_on} = \frac{\lambda_1}{\lambda_1 + \lambda_2} \times \frac{x_2}{x_2 + y_2} + \frac{\lambda_2}{\lambda_1 + \lambda_2} \times \frac{x_1}{x_1 + y_1}$$

Next we consider the case that only one flow is during On period. Given source 1 is during On period, the probability of seeing source 2 during Off period is:

$$p_{s1_on} = \frac{\lambda_1}{\lambda_1 + \lambda_2} \times \frac{y_2}{x_2 + y_2}$$

Given source 2 is during On period, the probability of seeing source 1 during Off period is:

$$p_{s2_on} = \frac{\lambda_2}{\lambda_1 + \lambda_2} \times \frac{y_1}{x_1 + y_1}$$

Similarly, due to the tail effect, we have to consider the idle time compensation. In the 2-flows case, the compensation will be:

$$\frac{y_1}{x_1 + y_1} \times \frac{y_2}{x_2 + y_2} \times compensate_d_delay$$

Where the *compensated_delay* is calculated by formula (5)

Then the delay is estimated as follows:

$$p_{both_on} \times d_{both_on} + p_{s1_on} \times d_{s1_on} + p_{s2_on} \times d_{s2_on} + idle_time_compensation$$

d_{both_on} is the delay of both traffic during On period, it is calculated by using the D/D/1/K model with arrival rate $\lambda_1 + \lambda_2$; d_{s1_on} is the delay of only source 1 during On period and is calculated by using D/D/1/K model with arrival rate λ_1 ; d_{s2_on} is the delay of only source 2 during On period and is calculated by using D/D/1/K model with arrival rate λ_2 .

Now, in order to make sure that our estimation is correct, we make a contrast between our model and ns2. Here we assume there are two Exponential On/Off flows entering one RED queue where the parameters are $min_{th}=10$, $max_{th}=30$ and $max_p=0.1$. The mean time of On period and Off period of each traffic are all the same—500ms. Then we list the result of the experiment as the following table:

Arrival rates	Model type	Delay	Difference
S1 = 6Mbps S2 = 6Mbps	NS	0.0143355	Delay < 5%
	Exponential On/Off model	0.0149850	
S1 = 7Mbps S2 = 7Mbps	NS	0.0140624	Delay < 7%
	Exponential On/Off model	0.0152930	
S1 = 8Mbps S2 = 8Mbps	NS	0.0154755	Delay < 2 %
	Exponential On/Off model	0.0153820	
S1 = 9Mbps S2 = 9Mbps	NS	0.0154918	Delay < 1 %
	Exponential On/Off model	0.0154435	

Table 4. Comparison with Exponential On/Off traffics and NS2

4.QoS mapping

This section we will discuss the QoS mapping that is from 3G network to DiffServ Domain. The UMTS had defined 4 different service classes — Conversational, Streaming, Interactive and Background. Each class has its corresponding application and QoS requirements such as the Table 1. And the DiffServ domain define 3 different service classes that are EF, AF and BE respectively. We will then use the queueing model discussed in Section 3 to investigate the

influence of the mapping from 3G to DiffServ Domain.

4.1 Homogeneous QoS Mapping

In Section 3, we have proposed 3 different Queueing Model that are corresponding to different traffic types. Now we can use that model to implement the homogeneous mapping. Suppose there are 6 queues in the Ingress of DiffServ Domain, 1 for EF class, 4 for AF sub-classes (from AF1 to AF4) and 1 for Best-Effort. The scheduler is the Weighted Round Robin, so each queue has a different service rate.

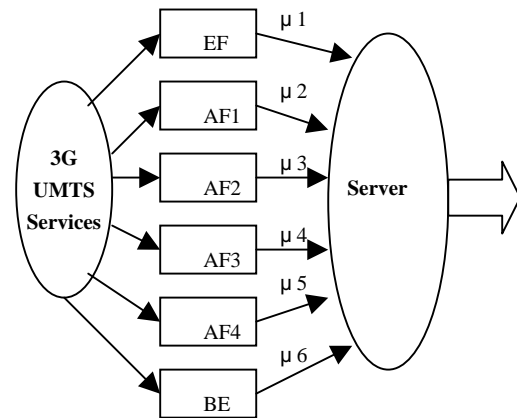


Figure 4. Mapping Architecture

EF is a kind of resource-reservation service class and the traffic type is usually CBR, so the queue model will be a D/D/1/K with FCFS queue. Packets are enqueued when the buffer has enough space and are dropped when the buffer is full.

AF provides the relative QoS instead of absolute QoS. It means that AF1 will receive lower drop precedence than AF2 to AF4, and AF2 will receive lower drop precedence than AF3 and AF4. AF service class is implemented by some queueing mechanism to ensure the relative QoS. Here we use RED (Random Early Detection) queueing mechanism to implement the 4 sub classes of AF.

As we mentioned before, RED queue has 3 main parameters— min_{th} , max_{th} and max_p . According to different consideration, AF1 to

AF4 will have different parameter settings. Then the 4 sub classes of AF will have different service rate (μ_k , where the k is from 2 to 5 in Figure 4) and drop function.

In order to make the homogeneous mapping, we implement the AF1 queue as the D/D/1/K model that the arrival traffic type is CBR; AF2 queue as the Exponential ON/OFF model that the arrival traffic type is Exponential On/Off; AF3 and AF4 queue as the M/D/1/K model that the arrival traffic type is Poisson. Then we model the AF queues as Table 5

AF	Traffic Type
AF1	CBR
AF2	CBR
AF3	Exponential On/Off
AF4	Poisson

Table 5. Homogeneous Queue Type

Last, the Best-Effort class can be modeled as M/G/1 model. The arrival traffic type is Poisson and has loose QoS requirement. The service rate is a general distribution because it varies depends on the situation of other queue. When some AF queue are empty, the service rate of BE queue will be larger than that when most of the AF queues are busy.

4.2 Admission Policy for the QoS

mapping

Now we consider the admission policy for the QoS mapping from UMTS service classes to DiffServ service classes. Because the Conversational and Streaming class have strict QoS requirement, we will map it to EF class if the remaining bandwidth is available. If the EF bandwidth is not available, we could also map it to AF1 or AF2 classes. But, we must use D/D/1/K model to estimate the delay and loss to make sure that this mapping will still conform to the QoS requirement of UMTS. Interactive and Background classes

have loose QoS requirement and the reservation type are dynamic (Table 1). We can easily map it to AF class without any resource-reservation. Here we let AF3 and AF4 be the corresponding queue to Interactive and Background. By using the M/D/1 model and Exponential On/Off model, we can estimate its delay and loss and use this information to perform the admission policy.

So here we propose our admission policy for the QoS mapping as follows:

If arrival traffic type is “*Conversational*” or “*Streaming*” then

If **EF** bandwidth is available then

Map this session to **EF**

Elseif **EF** bandwidth is unavailable then

Use **D/D/1/K** model to estimate

the delay, loss and jitter of **AF1** queue

If the estimating QoS correspond to UMTS then

Map this session to **AF1**

Elseif not corresponding

Use **D/D/1/K** model to estimate the delay, loss and jitter of **AF2** queue

If the estimating QoS correspond to UMTS then

Map this session to **AF2**

Else

Reject this session

If arrival traffic type is “*Interactive*” then

Use **M/D/1/K** model to estimate the delay, loss and jitter of **AF3** queue

If the estimating QoS correspond to UMTS then

Map this session to **AF3**

Else

Use **M/D/1/K** model to estimate the delay, loss and jitter of **AF4** queue

If the estimating QoS correspond to UMTS then
Map this session to AF4
 Else
 Reject this session
 If arrival traffic type is “Background” then
 Use M/D/1/K model to estimate the delay, loss and jitter of AF4 queue
 If the estimating QoS correspond to UMTS then
Map this session to AF4
 Else
Map this session to Best-Effort

4.3 Simulation

This session we make the simulation with our mapping policy. The simulation environment is that total bandwidth for Ingress is 40Mb, and the bandwidth distribution is EF—10Mb, AF—20Mb and Best-Effort for 10Mb.

Here we suppose there are n1 Conversational traffics, n2 Streaming traffics, n3 Interactive traffics and n4 Background traffics. The QoS requirements of each service class are as table 5

Service	Data Rate	Delay	Delay Variation	Reliability
Conversational	4~25 kbps	<150 ms	<1 ms	< 3 %
Streaming	32~384 kbps	<10 sec	<1 ms	< 1 %
Interactive	4~13 kbps	<1 sec	<1 ms	< 3 %
Background	Not Defined	Not Defined	Not Defined	~ 0 %

Table 6. QoS Requirement Parameters [18]

We let a simple mapping policy is that directly map the Conversational and Streaming to EF, Interactive to AF and Background to Best-Effort. This simple mapping (or we call it default mapping) is a straightforward mapping without QoS consideration. Thus, we make the contrast between the simple mapping and our homogeneous mapping policy.

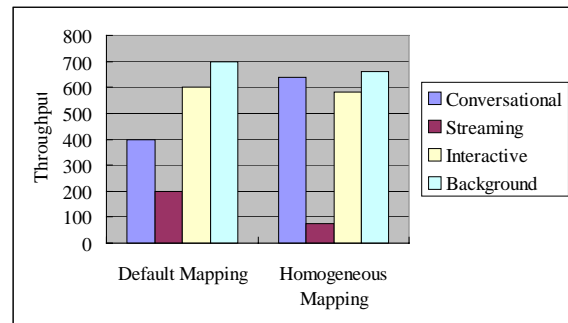


Figure 5. Throughput comparison between 2 mapping policies

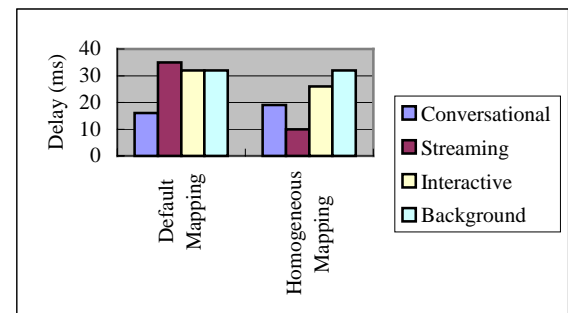


Figure 6. Delay comparison between 2 mapping policies

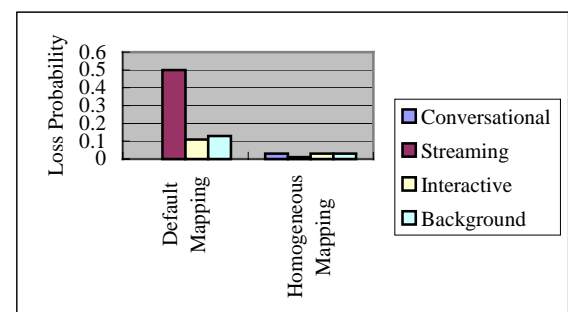


Figure 7. Delay comparison between 2 mapping policies.

From Figure 5 to 7, although the homogeneous mapping we proposed degrades the throughput, it extremely promotes the QoS for each service class. Thus, our mapping policy is based on the QoS achievement instead of throughput.

5. QoS Adaptation

In section 4, we have proposed the QoS admission policy for mapping and evaluate the performance of each queue model. A service provider should guarantee the service quality for all customers who have been admitted to the system. Thus, when the service provider's system and network conditions are changing, service provider must be able to dynamically adapt the system in order to guarantee the system quality.

In this section, we propose the QoS adaptation function and evaluate the influence of adaptation. Our adaptation is focus on the class adaptation instead of flow adaptation. Because DiffServ uses RED queue to implement the service provisioning, we could adapt the RED parameters to perform the QoS adaptation. The adaptation aspects could be delay or loss probability and this is a trade-off issue. Raising the loss probability could obtain better delay performance and increasing the max threshold could get lower loss probability but higher delay.

5.1 QoS Adaptation Function

This section we describe a QoS adaptation function based on the proposed queue model. It is composed of a monitoring function, an assessment function, and a control function. It is shown as Figure 8.

The monitoring function plays the role of monitoring the performance of the RED queue and network resource status. The assessment function decides whether a QoS violation occurs or QoS restoration is required. If required, the control function adjusts the parameters of the RED queue to guarantee the QoS of each queue.

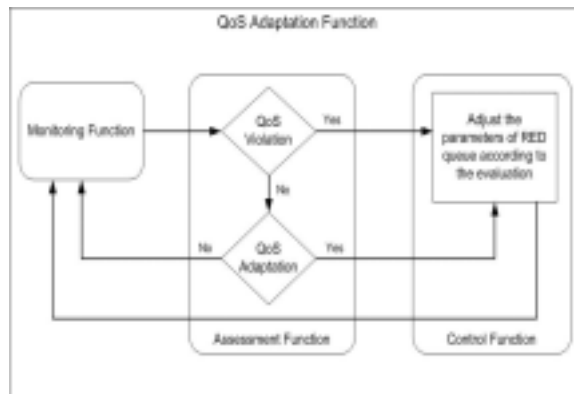


Figure 8. QoS adaptation functional description

The QoS adaptation function can be processed as follows:

Step 1: Initialization

Step 2: Performance monitoring

Step 3: Current state assessment

a) If QoS violation occurs, then go to Step 4

b) If QoS Adaptation is required, then go to Step 4

c) Go to **Step 2**

Step 4: QoS adaptation

a) Determine the parameters of RED queue

b) Adjust the parameters according to the adaptation evaluation

c) Go to **Step 2**

Next section we will propose the adaptation evaluation. The adaptation evaluation can be used for service provider to determine how to adjust the RED queue and the influence of changing those parameters.

5.2 QoS Adaptation Evaluation

Basically, RED queue uses 3 parameters that are \max_{th} , \min_{th} , and \max_p . Augmenting the value of \max_{th} could reduce the loss probability but increase the delay. So we firstly investigate the influence of changing the value of \max_{th} .

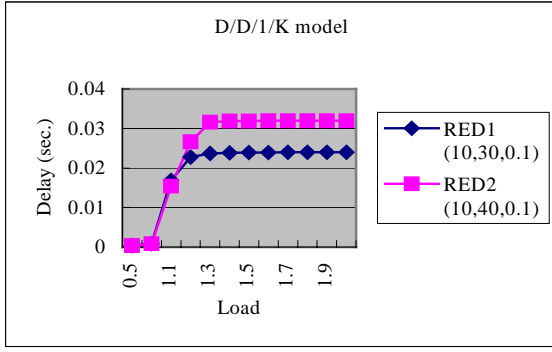


Figure 9. *D/D/1/K model with different parameter of max_{th}*

Now we consider the situation of changing the value— max_p :

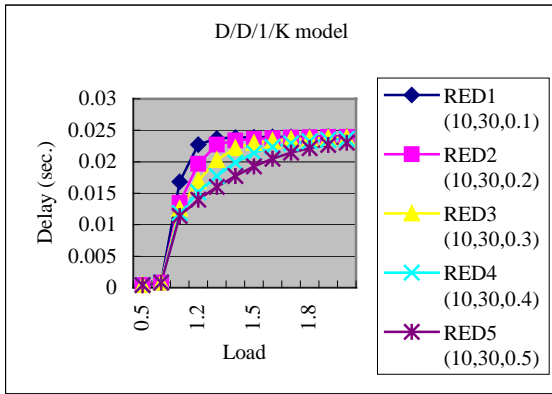


Figure 10. *D/D/1/K model with different parameter of max_p*

From these evaluation, we can see that when the system load is bigger than 1, adjusting the parameter of RED queue could get high performance of delay and the drop probability is not increase too much. For instance, we take a look at the D/D/1/K model with system load is 1.2. The variation of delay and loss are shown as Table 7.

	RED (10,30,0.1)	RED2 (10,30,0.5)	Improvement
Avg. delay	0.022735	0.013967	38 %
Drop Pkts	10000	10024	2.4 %

Table 7. *Contrast between delay and loss*

In Table 7, we adjust the parameter max_p from 0.1 to 0.5, and the average delay

improves 38% while the dropped packets just increase 2.4%. This is because raising the value of parameter max_p makes the number of early dropped packets increase. So this adaptation can be used to dynamically adjust the RED queue based on the different traffic types. For instance, if the arrival traffic type is Conversational or Streaming, which is more sensitive on the delay instead of loss, we could increase the parameter max_p to obtain better delay performance. Similarly, when the arrival traffic type is Background, we could greaten the maximum threshold— max_{th} to decrease the loss. By using this concept, we could define the QoS mapping as the following table:

Service Type	Parameter needed to be Adapted
Conversational	Increase max_p
Streaming	Increase max_p
Interactive	Decrease max_{th} and Increase max_{th}
Background	Decrease max_{th} and Increase max_{th}

Table 8. *Adaptation Method*

Thus, our adaptation evaluation provides an efficient way to determine how to adjust the parameters of RED queues.

6. Conclusions and Future Work

While most of the work on Quality of Service has focused on specifying the service type and definition, our work addresses the issues for defining the mapping policy that performs the mapping between 2 different network domains which both has its own QoS definitions. The 2 different network domains we discuss here are 3G and DiffServ. UMTS had defined the 4 different QoS service classes for 3G and DiffServ had 3 fundamental PHB types for implementing the QoS.

In this paper, we firstly propose the traffic model for modeling the arrival traffic to estimate the QoS parameters—delay, loss and jitter. According to different traffic types, we propose different corresponding models that are D/D/1, M/D/1 and Exponential

On/Off models. Then from these models, we address the homogeneous mapping for admission policy. The homogeneous mapping policy estimates the delay and loss and decides how to map or reject. It promotes the QoS achievement but degrades the throughput. This mapping policy just provides the concept for performing the mapping based on different aspects.

Future works will first focus on the heterogeneous mapping policy. This will cause the complicated situation for performing estimation and traffic modeling. Then we may also think about the QoS adaptation. The adaptation could be used to re-map the current session or adjust the bandwidth distribution for each service type. These issues are needed to research. And, we believe the mapping policy will make the network more efficient.

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