A Distributed Table-Driven Route Selection Scheme for Establishing Real-Time Video Channels

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Abstract — *To guarantee the delivery of real-time messages before their deadline, a real-time connection or channel must be established before the transmission of any real-time messages. During this channel-establishment phase, one must first select a route between the source and destination of this channel and then reserve sufficient resources along this route so that the worst-case end-to-end delay over the selected route may not exceed the user-specified delay bound.*

We propose a table-driven distributed route-selection scheme that is guaranteed to find a "qualified" route, if any, that meets the performance requirement of the requested channel without compromising any of the existing guarantees. The proposed scheme uses the Bellman-Ford shortest path algorithm to build real-time delay tables, and hence, can solve the route-selection problem by a simple table look-up. Several examples are presented to demonstrate the effectiveness of the proposed distributed route-selection scheme.

1 Introduction

An increasing number of applications such as interactive video and computer-integrated manufacturing require real-time networking services. Among the several real-time communication protocols proposed thus far to meet this requirement, the communication abstraction called the "real-time channel" [3] has received considerable attention due to its conceptual simplicity. A real-time channel is a unidirectional virtual circuit which, once established, is guaranteed to meet user-specified performance requirements as long as the user does not violate his "contract" terms [3].

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There are two distinct phases with the realization of a real-time channel: off-line channel establishment and run-time message scheduling. During the channel-establishment phase the system has to select a route between the source and destination of the channel along which sufficient resources can be reserved to meet the user-specified delay and buffer requirements. Although several channel-establishment schemes have been proposed in the literature [3,4], very few of them have addressed explicitly the issue of selecting a route between the source and destination of a channel, despite its importance to channel establishment.

Since the number of possible routes between two communicating peers in a multi-hop network could be large, selecting a route for each real-time channel is potentially a time-consuming task. It is therefore very important to develop an efficient scheme that is guaranteed to find a "qualified" route, if any, for each requested real-time channel. Given the worst-case anticipated traffic over a real-time channel, a "qualified" route for this real-time channel is defined to be the one that can meet the user-specified end-to-end delay requirement without compromising any of the existing guarantees. The service provider (the network operating system in our case) must also be able to reject a channel-establishment request as soon as possible if no qualified routes are available for the requested channel.

There are basically two approaches to the route-selection problem: centralized or distributed. Most existing channel-establishment schemes are centralized [4]. They simply assume the existence of a global network manager which maintains the information about all the established real-time channels, the topology and resource distribution & commitment of the underlying network, and can thus select an appropriate route for each real-time channel requested. In such a centralized scheme, all of real-time channel-establishment requests require the network manager's approval. Although using a centralized scheme one can devise efficient algorithms for the network manager to select qualified routes, there are two serious problems with this scheme. First, the network manager is likely to be a performance bottleneck, since it must handle all channel establishment and disconnection requests. Second, the system is susceptible to a single-point failure. In contrast with the centralized approach, distributed route-selection can avoid performance and reliability bottlenecks. However, since there could be many possible routes between two communicating peers, it may be too time-consuming to search for all possible routes and perform an admission test on each candidate route during the channel-establishment phase. On the other hand, if we only test a small number of routes, we may not find a qualified route even if there exists one.

In this paper, we propose a distributed table-driven route-selection scheme which takes advantage of the highly-periodic nature of interactive video and maintains a *real-time delay table* (or simply *delay table*) at each node so that a route for a real-time video channel may be selected by a simple table look-up.

The paper is organized as follows. Section 2 states the problem of finding a qualified route for each real-time video channel. Our proposed solution is presented in Section 3. Section 4 illustrates the effectiveness of our solution. The paper concludes with Section 5.

2 Problem Statement

There are two simple-minded approaches to the distributed route-selection problem: (1) sequential search of all possible routes one by one, or K routes at a time; (2) parallel search of all possible routes, i.e., sending multiple copies of an establishment request through all possible routes, "conditionally" reserving resources and performing admission tests on all of them. The second approach is practically infeasible due to its excessive operational overhead. The first approach, on the other hand, could be very time-consuming for the complete search of all possible routes, and its operational overhead is proportional to K.

Considering the advantages and disadvantages of these two approaches, we have developed in [6] an efficient scheme for selecting a route for each channel-establishment request. Although this scheme is guaranteed to find a qualified route, if any, for a *single* channel-establishment request, it cannot guarantee the qualified routes to be found for multiple (near) *simultaneous* channel-establishment requests due to its reliance on the over-estimation of link delays. Since this scheme is intended for channels with *general* performance requirements and traffic patterns, its performance could be improved significantly if real-time traffic is limited to certain types. Specifically, we will in this paper consider the real-time traffic of interactive video applications. Interactive video applications usually generate frames at some fixed rate[1] and resolution which are both specified according to industry standards. For example, 30 frames per second is the frame rate for live interactive video and the MPEG Video Simulation Model Three (SM3) suggests 352 by 288 pixels per frame for achieving video tape quality [5]. Note that a standardized resolution implies a standardized maximum-frame size. Since video applications of our interest require only a small set of combinations of frame-generation rates and maximum-frame sizes, by exchanging and maintaining real-time traffic information among the nodes, the system may be able to prepare for channel establishment even *before* receiving a establishment request. Under this setting, we will develop a scheme which builds and maintains a delay table on each node so that the route-selection problem can be solved by a simple table look-up at the source node.

3 The Proposed Solution Approach

We first describe the environment and the assumption under which our distributed route-selection scheme will be developed. The underlying network is an arbitrary point-to-point network. As in [2, 4, 6], the generation of real-time messages is assumed to be governed by a linear-bounded arrival process that is characterized by three parameters: maximum message size S_{max} (bytes), maximum message rate R_{max} (messages/second), and maximum burst size B_{max} (messages). In the linear bounded model, there are two restrictions: (1) the number of messages generated in any time interval of length t does not exceed $B_{max} + tR_{max}$; (2) the length of each message is bounded by S_{max}. Based on this message arrival model, the authors of [4] proposed a scheme to estimate the worst-case delay on each link and a run-time scheduling algorithm for real-time messages. By adding the worst-case delays of all links that a channel runs through, one can calculate the worst-case end-to-end delivery delay. This end-to-end delay is then compared against the user-specified end-to-end delay bound for the requested channel and the system can decide whether to accept/reject the corresponding channel request. Note that these schemes have been developed under the assumption that a proper route for the requested channel was already available.

Besides the linear bounded model, we further assume

[1] allowing jitters

that the number of possible combinations of frame-generation rates and maximum-frame sizes is small, since we are only interested in standardized interactive video applications. Based on the link delay calculated with the delay-estimation method in [4] and the above assumptions, we will develop a scheme which builds real-time channel delay tables on each node so that a qualified route may be found by a simple table look-up.

3.1 Link-Delay Estimation

Since real-time messages are given priority over non real-time ones, we will ignore the effects of non real-time traffic in the rest of the paper unless stated otherwise. We will thus assess the delay of a link based only on real-time traffic. Since the algorithm in [4] will be used to estimate link delays, we will briefly introduce it first.

The goal of the algorithm in [4] is to compute the minimum worst-case response time (MWRT) on a link of each candidate route for a new real-time channel to be added without compromising the performance guarantee of any of the existing channels on the link. Let $\{M_i = (C_i, p_i, d_i), i = 1, \ldots, k\}$ be the set of k existing channels on a link, where C_i is the maximum time required to transmit a message of channel M_i on the link, p_i is the minimum message inter-arrival time in M_i, and d_i the maximum permissible delay assigned to M_i on this link, or *link deadline*. Note that the inequality $d_i \leq p_i$ must hold for the algorithm in [4] to work correctly. Given a new channel $M_{k+1} = (C_{k+1}, p_{k+1})$ to be established, the algorithm can compute the the MWRT of M_{k+1}, r_{k+1}, on link ℓ based on the traffic-generation characteristics (C and p) of the channel, when C, p and d for all existing channels on link ℓ are available. The algorithm statically assigns priority to each real-time channel (in ascending order of d values) to calculate the MWRT for M_{k+1}, but uses an Earliest-Due-Date (EDD) algorithm for run-time scheduling.

The method in [4] does *not* include those channels pending for final confirmation in the calculation of MWRT for the new channel-establishment request, but we will treat pending channels differently in two situations. First, during the channel-establishment phase, we will include the load of pending channels in the calculation of MWRTs as if they had already been established. However, the load of pending channels will *not* be included in the real-time delay tables, i.e., we do not include pending channels in the calculation of MWRTs which are used to build real-time delay tables.

Including pending channels in the calculation of MWRTs can simplify the channel-establishment phase, since the MWRTs remain valid when the confirmation message travels back from the destination to the source. However, inclusion of these pending channels in the link-delay estimation will sometimes make the MWRT larger than it actually would be if some of them are rejected later. This over-estimation of MWRT may result in incorrect rejections of channel-establishment requests. Fortunately, the over-estimation problem occurs only when two requests initiate at about the same time and their least-MWRT paths share one or more links. In order to avoid any possible confusion, "existing channels" will henceforth mean *both* established and pending channels. Determination of each channel's MWRT on a link will be referred to as *link-delay estimation*.

3.2 Building Real-Time Delay Tables

Based on the above definition of link delay, we can apply the Bellman-Ford algorithm [1] and a loop-free version of the APARNET's previous routing strategy (APRS) [7] to build real-time delay tables on each node. As mentioned earlier, the MWRT used to construct real-time delay tables does not include the load of pending channels for two reasons. (1) The maximum permissible delay for a link after a final confirmation is likely to be greater than the MWRT computed during the resource-reservation phase. (2) The time between making resource reservation and receiving a final confirmation is usually small, e.g., it could be the time needed for the round-trip from a node on the route under test to the destination. Thus, if we want to include the load of pending channels in the real-time delay tables, the table entries may have to be modified twice in a short period of time.

When only a small set of standardized combinations of frame-generation rates and maximum-frame sizes needs to be considered, each node in the network can build a loop-free table based on the MWRTs computed according to a pair of maximum-frame size (S_{max} or C) and frame-generation rate (p). All real-time channels that can be specified by the same pair (S_{max}, p) are said to be in the same *class*. A node will compute the MWRT on each of its outgoing links as the minimum feasible delay of the corresponding link for *each class* of real-time channels. These MWRTs will be stored in a table, TM, which can be indexed by its neighbors' addresses and has only one field, r, for each class of channels, representing their MWRTs over the corresponding link by considering only those channels already established.

Since the loop-free APRS will be used to exchange delay information and maintain real-time delay tables, we briefly describe the original APRS. In APRS, each node collects and maintains the information about the minimum delays to all the other nodes via each of its neighbors. For every destination–neighbor pair, the information is kept in a 3-tuple form (*destination*, *neighbor*, *delay*). Other information may be needed for the loop-free APRS, but we will not discuss this issue here. (See [7] for a detailed account of this.) These 3-tuples are divided into groups based on the destination node. Within a group, they are listed in the ascending order of delay. The first entry of each group (the minimum-delay entry to the corresponding destination node) is then used to build a routing table.

Each node periodically exchanges a routing message with its neighbors which contains the node's current routing table. After receiving routing messages from its neighbors, each node will update its own routing table based on the information carried in the routing messages and the status of its own outgoing links. As we shall see, the real-time delay table is built in the same way as the routing tables except for the following two differences. (1) Only real-time traffic is considered when building real-time delay tables. (APRS's routing tables were built by considering *all* traffic.) (2) Real-time delay tables are updated only when a new real-time channel is established or an existing real-time channel is closed. Thus, a node sends "routing" messages to its neighbors only when its set of real-time channels changes. These "routing" messages are called *real-time routing* messages. Using an example, we will show how to build real-time delay tables.

Example 1: The class of real-time channels under con-

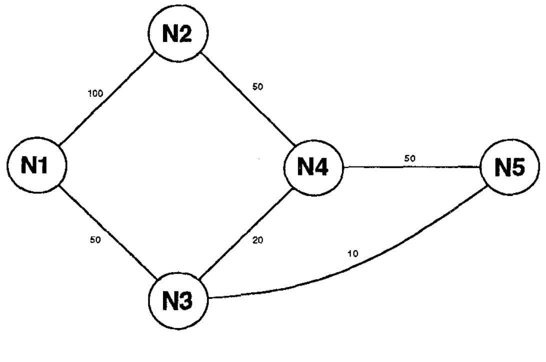

Figure 1: Example 1

sideration is specified by $S_{max} = 100$ Kbps and $p = 33$ ms. Fig. 1 shows the network used for this example. The number on each link represents its transmission speed. Since initially there are no existing real-time channels, if a channel-establishment request of this class is received, the highest priority will be given to the requested channel. The MWRT of this class for each link is thus equal

N_1	N_2	N_3	N_4	N_5
D,N,d	D,N,d	D,N,d	D,N,d	D,N,d
2,2,1	1,1,1	1,1,2	2,2,2	3,3,10
2,3,∞	1,4,∞	1,4,∞	2,3,∞	3,4,∞
3,3,2	4,4,2	1,5,∞	2,5,∞	4,4,2
3,2,∞	4,1,∞	4,4,5	3,3,5	4,3,∞
		4,1,∞	3,2,∞	
		4,5,∞	3,5,∞	
		5,5,10	5,5,2	
		5,1,∞	5,2,∞	
		5,4,∞	5,3,∞	

Table 1: Initial real-time delay tables, where D = Destination, N = Neighbor, d = delay.

N_1	N_2	N_3	N_4	N_5
D,N,d	D,N,d	D,N,d	D,N,d	D,N,d
2,2,1	1,1,1	1,1,2	2,2,2	3,4,7
2,3,9	1,4,9	1,4,8	2,3,8	3,3,10
3,3,2	4,4,2	1,5,15	2,5,15	4,4,2
3,2,8	4,1,8	4,4,5	3,3,5	4,3,15
4,2,3	3,1,3	4,1,5	3,2,5	1,4,5
4,3,7	3,4,7	4,5,12	3,5,12	1,3,12
5,2,5	5,4,4	5,4,7	5,5,2	2,4,4
5,3,9	5,1,10	5,1,7	5,3,15	2,3,13
		5,5,10	5,2,∞	
		2,1,3	1,2,3	
		2,4,7	1,3,7	
		2,5,14	1,5,14	

Table 2: Steady-state real-time delay tables.

to the maximum service time (C) of the class and can be computed as: link $N_1 \leftrightarrow N_2$: 100/100=1 ms; link $N_1 \leftrightarrow N_3$: 100/50=2 ms; link $N_2 \leftrightarrow N_4$: 100/50=2 ms; link $N_3 \leftrightarrow N_4$: 100/20=5 ms; link $N_3 \leftrightarrow N_5$: 100/10=10 ms; and link $N_4 \leftrightarrow N_5$: 100/50=2 ms.

Table 1 shows the real-time delay tables for all 5 nodes in the very beginning. Initially, a node can reach only its neighbors since information about the other nodes is not yet available to the node. The label ∞ in an entry represents the case when either the destination cannot be reached via the corresponding neighbor or the path via this neighbor is not loop-free. The least-MWRT path to each destination known so far is used to construct a real-time routing message. Each entry of the message is a 2-tuple, (*destination*, *delay*), where *destination* is not the neighbor to which this message will be sent. For example, the message from N_3 to N_4 will contain two entries: $(N_1, 2)$ and $(N_5, 10)$.

Since the real-time routing messages are not sent periodically (i.e., the updating procedures of real-time delay tables are not synchronized among nodes), it is not certain what the "next" state of the real-time delay tables will be. However, the "steady-state" of real-time delay tables depend only on the currently-anticipated real-time traffic of the established real-time channels. Thus, after all nodes stop sending real-time routing messages (before the next channel establishment or channel closing), the real-time delay tables at that moment can be determined from the current real-time traffic load, regardless of intermediate states. Table 2 shows the steady state of real-time delay tables before the arrival of any real-time channel-establishment request or the closure of any established channel. □

3.3 The Route-Selection Algorithm

In addition to the real-time delay tables, each node has to maintain another set of tables — called the *tables of existing channels* (TEXCs) — for the existing channels, one for each outgoing link. Each entry of a TEXC represents a real-time channel which goes through the corresponding link and consists of the following four fields: (1) channel identifier (ID) which uniquely identifies a real-time channel; (2) class (*class*) of the channel; (3) status (*status*) of the channel, established (1) or pending (0); (4) the maximum permissible link delay (d) for the channel. To ensure the uniqueness of channel ID, each ID consists of two parts, the source-node address and a number (unique within the source node).

When the source node wishes to establish a real-time channel to another node, say B, it will try to find the current least-MWRT route by considering the traffic of all existing channels. The source node will send a real-time channel-request message (*Req*) to the next node on the least-MWRT route, which contains a channel ID, the destination address (*dest*), the channel class (*class*), the end-to-end delay bound D, the path (*path*), the total number of hops (*hops*) traveled thus far, and the accumulated delay d^a. Initially, d^a is set to the MWRT of the corresponding outgoing link, *path* is set to the source node, and *hops* is set to 1. Although *hops* is included in the request message, it can be omitted in a real implementation because the information carried in *hops* can be derived from *path*.

```
Procedure rcv_req
 If ($Req.dest = A$) then {reply_req; return; }
 for ($i = 1$ to number_of_entry($RTDT[Req.dest]$)) {
  If ($Req.d^a + RTDT[Req.dest][i].d > Req.D$)
  then {send_reply(reject); return; }
  $nextnode := RTDT[Req.dest][i].N$;
  If (no pending channel in $TEXC[nextnode]$) then
   {insert_req($TM[nextnode]$); forward_req(nextnode);
   return;}
  % compute MWRT with established+pending channels.
  $r :=$ compute_MWRT(nextnode, 1);
  If ($Req.d^a + RTDT[Req.dest][i].d - TM[nextnode] + r$
   $\leq Req.D$) then {insert_req($r$); forward_req(nextnode);
   return;} }
 send_reply(reject);
```

Figure 2: Procedure of processing a request.

Fig. 2 describes the procedure of handling a channel-establishment request after node A receives the request. This procedure can also be applied to the source node by setting $d^a := 0$, *hops* $:= 0$, and *path* to an empty string. Procedure *rcv_req* uses the destination, B, as an index to the real-time delay table and searches through all routes whose delays to B are $\leq Req.D - Req.d^a$. The **for** loop and the first **if** statement in the loop serve as this function.

After passing the first **if** statement in the loop, if there is no pending channel (the second **if** in the loop), this entry is selected and appropriate actions will be taken by calling Procedure *insert_req* and *forward_req*. Otherwise, we have to re-compute the MWRT ($= r$) for the corresponding outgoing link. If the increase of MWRT due to the pending channel ($-TM[nextnode] + r$ part in the third **if** statement) doesn't make the delay to the destination greater than $Req.D - Req.d^a$, this entry can be selected as the channel's route. The maximum permissible delay on this link (d field in TEXC) is set to r, instead of obtaining it directly from $TM[nextnode]$. RTDT in *rcv_req* denotes real-time delay tables.

```
Procedure insert_req(r)
 $TEXC.ID := Req.ID$;
 $TEXC(Req.ID).class := Req.class$;
 $TEXC(Req.ID).status := pending$;
 $TEXC(Req.ID).d := r$;
Procedure forward_req(nextnode)
 $Req.d^a := Req.d^a + TEXC(Req.ID).d$;
 $Req.hops := Req.hops + 1$;  % concatenate A and Req.path.
 $Req.path := A \cdot Req.path$;  % other fields remain unchanged.
 forward this request message to nextnode.
```

Figure 3: Procedures of inserting/forwarding a request.

Fig. 3 describes the procedures of inserting a new (pending) channel to TEXC and forwarding a request. As can be seen from these procedures, most fields are directly copied from the establishment-request message to TEXC and the forwarding message.

```
Procedure reply_req
 If (the request is accepted) then send_reply(accept);
 else send_reply(reject);
Procedure send_reply(accept)
 $nextnode := head(Req.path)$;
 $Reply.ID := Req.ID$;
 $Reply.flag := accept$;
 $Reply.diff := (Req.D - Req.d^a)/Req.hops$;
 $Reply.path := tail(Req.path)$;
 send Reply to nextnode.
```

Figure 4: Procedure of processing a channel-establishment request at the destination.

Fig. 4 shows a destination's operations after receiving a channel request. Since the d^a field of a channel-establishment request represents the sum of MWRTs of all links in the path from the source to destination, the user-specified end-to-end delay bound, D, may be larger than d^a, i.e., we are allowed to spend more time

```
Procedure forward_reply
 If ($Reply.flag = reject$) then delete $TEXC(Reply.ID)$;
 else { $TEXC(Reply.ID).status := established$;
  $TEXC(Reply.ID).d := TEXC(Reply.ID).d$
    $+ Reply.diff$;
  insert this entry in the ascending order of $d$ field.
  % update real-time delay tables if necessary
  % assume the message is sent/forwarded by node N.
  update(N); }
 $nextnode := head(Reply.path)$;
 $Reply.path := tail(Reply.path)$;
 forward this message to the next upstream node $nextnode$.
```

Figure 5: Procedure of handling reply messages.

than the corresponding MWRT when sending a message over each intermediate link. In such a case, the authors of [4,6] proposed that $D - d^a$ should be divided evenly into *hops* parts by the destination node and distributed to all links along the path. The deadline of a real-time message of this particular channel over an intermediate link is the channel's MWRT of that link plus $diff = (D - d^a)/hops$. Note that one may also choose to divide $D - d^a$ in proportion to each link's MWRT. However, since this method may make the link-delay deadline unnecessarily small over a link which has small MWRT we will adopt the method proposed in [4,6].

The $diff$ will be included in the channel-establishment confirmation message (by procedure *send_reply*(*accept*)) from the destination to source via the same path the corresponding request message had traveled (but in the opposite direction). Let *Reply* denote a channel-establishment confirmation message which consists of four fields: ID, $flag$ (accept or reject), $diff$ and *path* (the remaining path back to the source node). Fig. 4 describes how a positive confirmation message is constructed (*send_reply*(*accept*)), and Fig. 5 shows the operations the intermediate nodes will perform when receiving a (positive or negative) reply message (*forward_reply*). Note that *head*(*list*) represents the first element of *list*, and *tail*(*list*) represents the remaining list after removing *head*(*list*) from *list*.

The operations in Procedure *update*(*node*) of Fig. 6 are necessary to keep these real-time delay tables and TMs up-to-date after a new channel is established or an existing channel is torn down. Basically, nodes which receive a positive reply to a channel request will recompute the MWRT by considering only those established already (including the one just accepted or excluding the one just closed). Based on this new MWRT, real-time routing tables and TM are updated and a new real-time routing message is generated and sent to all neighbor nodes. When a node receives a real-time routing message from a neighbor node, it will update its real-time delay tables.

```
Procedure update(node)
 $message := \phi$;
 for each class $class$ of real-time channels {
  % compute new MWRT with only established channels.
  $r := compute\_MWRT(node, 0)$;
  If ($r \neq TM[node]$) then {
   $update\_RTDT(node, r)$; $TM[node] := r$;
   $table := \cup_{i=all\ destinations}\{(i, RTDT[i][1].d)\}$; }
  $message := message \cup \{(class, table)\}$; }
 send $message$, to all neighbor nodes;
Procedure update_RTDT(node, r)
 for all destinations, $dest$ {
  $i := 1$;
  while ($i \leq number\_of\_neighbors$) {
   If ($RTDT[dest][i].N \neq node$) then $i := i + 1$;
   else {$RTDT[dest][i].d := RTDT[dest][i].d + r$
     $-TM[node]$; $break$; } } }
```

Figure 6: Procedure of updating real-time delay tables.

The operations necessary to keep real-time delay tables and TMs up-to-date during the channel-disconnect phase are straightforward. We require one of the two communicating peers to send a disconnect message (with ID) through the route of the real-time channel to the other communicating peer. All intermediate nodes and the source node will delete the corresponding entry from their TEXCs and *update* (Fig. 6) both TMs and real-time delay tables. Real-time routing messages may also be sent as discussed in Section 3.2.

3.4 Overhead Analysis

In this section, we analyze the storage overhead of our strategy, since the primary concern of using a table-driven route-selection algorithm is the size of the tables. The largest table used in the proposed strategy is the real-time delay table. All the other tables are much smaller than the real-time delay table. Thus, we will start with the derivation of an upper bound of the size of the real-time delay table.

Let N denote the number of nodes in the network and a denote the average number of neighbors of a node (or node degree). Then, for each class of real-time channels, the maximum size of all real-time tables in the network is $3aN(N-1)$. The first N represents the fact that there are N real-time tables per class in the network. $(N-1)a$ represents the number of entries for a real-time delay table and 3 is the number of records for each entry. Basically, the proposed strategy incurs $O(N^2)$ storage overhead for maintaining real-time delay tables.

The table of MWRTs is much smaller than the real-time delay tables. The size of all tables of MWRTs is only $2aN$, where N represents the number of such tables in the network, and there are a entries per table and 2 records per entry.

The size of the above two tables does not change regardless how many real-time channels are established and which routes they use. However, the size of TEXC does depend on the number of existing channels and the average number of hops for each channel. Let h denote the average number of hops for an existing channel and n denote the number of existing channels. Then the size of all TEXCs in the network can be specified as $4hn$, where 4 is the number of records per entry (representing an existing channel) in a TEXC. Note that h here is of $O(N)$ since our strategy chooses only loop-free routes.

Based on this analysis, unless the number of existing channels is extremely large (i.e., more than $O(N)$), our strategy incurs an $O(N^2)$ storage overhead for supporting each class of real-time channels.

4 Examples

Example 2: Based on Example 1, we want to establish a channel (ID=1:1) of this class (class 1) with an end-to-end delay bound D=32 ms from N_1 to N_5.

Link-bandwidth reservation:N_1 will use N_5 as the index to its real-time delay table (Table 2), and find the next node, N_2, on the least-MWRT path to N_5 with MWRT = 5 ms. Since $D > 5$ ms and there is no pending channel, N_1 will insert this channel into its TEXC for link $N_1 \rightarrow N_2$. Because $TM[N_2] = 1$ for this class, the entry representing this channel in TEXC can be set to (1:1,1,0,1). Req(1:1,N_5,1,32,N_1,1,1) can also be sent to the next node, N_2.

After receiving this request message, N_2 will call *rcv_req* to handle this request message and forward it to the next hop on the least-MWRT path to N_5. From Table 2, N_4 is the next hop and the delay is 4 ms. Since the accumulated delay (d^a) carried in the request is 1 ms, i.e., the end-to-end MWRT ($1+4=5$ ms) is not greater than $D = 32$ ms and there is no pending channel on link $N_2 \rightarrow N_4$, this channel request will be inserted into N_2's TEXC for link $N_2 \rightarrow N_4$. Because $TM[N_4] = 2$ (for class 1), the entry in TEXC will be set to (1:1,1,0,2). A request message, Req(1:1,N_5,1,32,N_2N_1,2,3), will also be forwarded to N_4.

N_4's operations are similar to N_2's. The entry inserted in N_4's TEXC (link $N_4 \rightarrow N_5$) for this request is (1:1,1,0,2), and the request message forwarded to N_5 is Req(1:1,N_5,1,32,$N_4N_2N_1$,3,5).

If the peer application at N_5 decides to accept this channel request, a positive reply message will be constructed and sent back to N_1 via the same path of the request message traveled but in the opposite direction. Using *send_reply*(*accept*), $diff$=(32-5)/3=9 and the positive reply message will be Reply(1:1,accept,9,N_2N_1). This reply message will then be sent to N_4

Channel-acceptance confirmation: After receiving the positive reply from N_5, N_4 will call *forward_reply* and update the entry in TEXC (link $N_4 \rightarrow N_5$) representing channel 1:1 to (1:1,1,1,11), where the third component indicates this channel to have been established, and the fourth component ($2+9=11$) shows the maximum permissible delay of this channel over link $N_4 \rightarrow N_5$. This positive reply (after removing *head*(*Reply.path*)) will then be forwarded to the next upstream node (N_2) specified by *head*(*Reply.path*).

Due to the establishment of channel 1:1, the MWRTs of channels (of all classes) over link $N_4 \rightarrow N_5$ have to be re-computed. However, by using the method described in Section 3.1, the MWRT for class-1 channels still remains to be 2 ms. Thus, N_4's real-time delay tables will remain unchanged and no real-time routing messages will be sent.

The operations performed by N_2 and N_1 are similar to N_4's. The TEXC for link $N_2 \rightarrow N_4$ will contain an entry (1:1,1,1,11) and the TEXC for link $N_1 \rightarrow N_2$ will contain an entry (1:1,1,1,10). All real-time delay tables and TMs remain unchanged.

If the application N_5 refuses to accept this request, a negative reply will be sent back via the same path. After receiving the negative reply, all nodes will delete the corresponding entry from their TEXCs and forward the reply message to the next upstream node specified by the *path* in the reply message. □

Example 3: Another class (class 2) of real-time channels specified by $S_{max} = 300$ Kbps and $p = 20$ ms will be used. Table 3 shows the tables of MWRTs for all 5 nodes after channel 1:1 is established (Example 2), and Table 4 shows the steady state of real-time delay tables for class-2 channels. We will establish a class-2 channel (ID = 1:2, destination = N_5, and $D = 30$ ms) and then show the change in both TMs and real-time delay tables after its establishment.

As in Example 2, the path $N_1N_2N_4N_5$ will be chosen because it is the least-MWRT path from N_1 to N_5. Thus, a request will be sent from N_1 to N_5 via this path and each node on this path will insert a corresponding entry into its own TEXC to reflect the existence of channel 1:2. The entries to be inserted into the TEXCs of N_1, N_2 and N_4 are: link $N_1 \rightarrow N_2$: (1:2,2,0,3), link $N_2 \rightarrow N_4$: (1:2,2,0,6), and link $N_4 \rightarrow N_5$: (1:2,2,0,6). If the peer application at the destination node accepts this channel-establishment request, N_5 will compute $diff = 30 - (3 + 6 + 6) = 5$ ms and send a positive Reply(1:2,1,5,N_2N_1) to N_4.

After receiving the positive reply to the request of establishing channel 1:2, N_4 will set the status of this channel to "established" and the maximum permissible delay to $6 + 5 = 11$ ms in the TEXC (link $N_4 \rightarrow N_5$). Then, N_4 will re-compute the MWRT of both classes over link $N_4 \rightarrow N_5$, including only the load of established channels. For class-2 channels, $TM[N_5]$ is then modified to 14. However, the MWRT of class-1 channels does not change. Thus, the real-time delay tables will be updated (Table 5) and real-time routing messages containing only the class-2 information will be sent to N_4's neighbors. For example, the message sent to N_2 will contain three entries: (N_5,14), (N_3,15), (N_1,21), and the message sent to N_3 will also contain three entries: (N_5,14), (N_2,6), and (N_1,9). Note that (N_1,21) in the message to N_2 is the result of loop-free routing.

The positive reply will be forwarded to N_2 then to N_1. N_2's operations are similar to N_4's. The MWRTs of both classes over link $N_2 \rightarrow N_4$ have to be re-computed and the real-time delay tables have to be updated accordingly. The MWRT of class 2 over link $N_2 \rightarrow N_4$ will increases to 14 and that of class 1 remains unchanged. So, $\{(N_5,28),(N_4,14),(N_3,29)\}$ will be sent to N_1, and $\{(N_1,3),(N_3,9),(N_5,38)\}$ will be sent to N_4 as the real-time routing messages for class-2 channels.

After receiving the confirmation of establishing channel 1:2, N_1 will also re-compute the MWRTs of both classes. Since both MWRTs do not change over link $N_1 \rightarrow N_2$, no further actions are necessary.

In addition to the positive reply messages, the real-time routing messages generated by N_4 and N_2 will also be used to update real-time routing tables by those nodes which receive the messages, as discussed in Section 3.2. Following the procedure of building real-time delay tables, Table 5 shows the steady-state real-time delay tables after channel 1:2 is established.

5 Conclusion

In this paper, we have proposed a table-driven distributed route-selection scheme which is guaranteed to find a qualified route, if any, for each real-time channel-establishment request. By equipping a real-time delay table with each node, our scheme can choose a route for each real-time channel requested by a table look-up.

N_1	N_2	N_3	N_4	N_5
N,MWRT	N,MWRT	N,MWRT	N,MWRT	N,MWRT
2,3	1,3	1,6	2,6	3,30
3,6	4,6	4,15	3,15	4,6
		5,30	5,6	

Table 3: Tables of MWRTs (N stands for Neighbor.)

N_1	N_2	N_3	N_4	N_5
D,N,d	D,N,d	D,N,d	D,N,d	D,N,d
2,2,3	1,1,3	1,1,6	2,2,6	3,4,21
2,3,27	1,4,27	1,4,24	2,3,24	3,3,30
3,3,6	4,4,6	1,5,45	2,5,45	4,4,6
3,2,24	4,1,24	4,4,15	3,3,15	4,3,45
4,2,9	3,1,9	4,1,15	3,2,15	1,4,15
4,3,21	3,4,21	4,5,36	3,5,36	1,3,36
5,2,15	5,4,12	5,4,21	5,5,6	2,4,12
5,3,27	5,1,30	5,1,21	5,3,45	2,3,39
		5,5,30	5,2,∞	
		2,1,9	1,2,9	
		2,4,21	1,3,21	
		2,5,42	1,5,42	

Table 4: Steady-state real-time delay tables for class-2 channels.

N_1	N_2	N_3	N_4	N_5
D,N,d	D,N,d	D,N,d	D,N,d	D,N,d
2,2,3	1,1,3	1,1,6	2,2,6	3,4,21
2,3,27	1,4,35	1,4,24	2,3,24	3,3,30
3,3,6	4,4,14	1,5,45	2,5,53	4,4,6
3,2,32	4,1,24	4,4,15	3,3,15	4,3,45
4,2,17	3,1,9	4,1,23	3,2,15	1,4,15
4,3,21	3,4,29	4,5,36	3,5,44	1,3,36
5,2,31	5,4,28	5,4,29	5,5,14	2,4,12
5,3,35	5,1,38	5,5,30	5,3,45	2,3,39
		5,1,37	5,2,∞	
		2,1,9	1,2,9	
		2,4,21	1,3,21	
		2,5,42	1,5,50	

Table 5: Steady-state real-time delay tables for class-2 channels after channel 1:2 isestablished.

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