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Abstract

The Controller Area Network (CAN) is a broadcast medium providing advanced features, which make it suitable for many real-time applications. Common application layer protocols designed for CAN (e.g. CAL, SDS, DeviceNet) exploit these features in order to provide reliable real-time communication. However, they do not provide a consistent global message ordering in certain fault situations, nor do they consider temporal properties of the application.

This paper presents a multicast protocol which supports timely delivery of messages and guarantees atomic order, under anticipated fault conditions. In order to support causal ordering of events, the paper provides a discussion on establishing Lamport's precedence relation between events by appropriate usage of real-time multcasts.

In a CAN-based distributed system, the restricted communication bandwidth constitutes a serious bottle-neck. Therefore an important feature of this multicast protocol is to achieve optimal protocol termination time while requiring minimum communication overhead.

\textit{Key words:} Real-time communication; fault-tolerance; multicasting; total order; CAN

1 Introduction

Future computer systems will, to a large extent, monitor and control real-world processes. This results in an inevitable demand for timeliness and reliability. Distributed systems which inherently provide extensibility and immunity against single failures, are an adequate architecture to cope with spatially distributed real world applications. Moreover, the availability of inexpensive, powerful micro-controllers promotes distributed solutions.

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Group communication is the basic mechanism to coordinate distributed activities. In order to control the competition and cooperation among distributed processes, and to allow for object replication, atomic delivery of multicast messages is inevitable. In this paper, the following definition of atomic multicast delivery is used.

**Definition 1** Atomic multicast delivery – Let $G$ be a multicast group, i.e. a set of objects which must receive a multicast message. Let $\text{del}_i(m)$ denote the delivery of a message $m$ to an object $i$ and $\rightarrow$ define the precedence relation. Then two messages $m$ and $m'$ are delivered atomically to the group $G$, if and only if the following holds:

$$\forall i, j : i, j \in G \land i, j \text{ non-faulty}$$

$$\Rightarrow (\text{del}_i(m) \rightarrow \text{del}_i(m') \Leftrightarrow (\text{del}_j(m) \rightarrow \text{del}_j(m'))$$

In other words, atomic delivery of multicast messages implies two properties:

- **Consistent Delivery**: if a multicast message is received by a non-faulty group member, then it is received by all non-faulty group members.
- **Consistent Ordering**: messages received by different non-faulty group members, are received in the same order.

Achieving the consistent delivery requires either a two-phased commit protocol with a considerable acknowledgment overhead (Babaoglu and Drummond, 1985; Birman and Joseph, 1987; Chang and Maxemchuk, 1984), or a mechanism to retransmit each message so many times that the probability of losing all copies is negligible (Cristian, 1990; Livani, 1998; Rufino et al., 1998).

Consensus on the ordering of the messages comes free with the two-phased commit protocol. But if the multiple transmission approach is applied, an additional mechanism for establishing a unique order of the messages must be used. (Schneider, 1990) has described an approach to achieve consensus on the message ordering among a group of receivers, using a time-dependent unique identifier of messages. An approach which exploits knowledge about the delivery deadlines of the messages, has been introduced by (Cristian et al., 1985). A similar mechanism has been proposed for embedded applications by (Zuberi and Shin, 1996). These approaches, however, rely on the timely and reliable delivery of all messages.

The ordering scheme presented in this paper considers also the late transmission of soft real-time messages, which is a timing failure of the message transmission, but does not lead to a system failure. As shown in section 3, it is possible to schedule the real-time communication in a CAN-bus system so that hard real-time messages are always transmitted timely while soft trans-
mission deadlines are missed in overload situations. In such a system it is not possible to guarantee a total delivery order among all hard and soft real-time messages.

To see the reason of this restriction consider a hard real-time message $H$, a soft real-time message $S$, and two nodes $N_1$ and $N_2$. Since no latest delivery time can be guaranteed for $S$, its consistent delivery must depend on the reception of a signal $\sigma$ from the bus (e.g. $S$ itself, a commit packet, or another packet depending on the protocol). Assume that $N_1$ receives $\sigma$ at time $t_1$, and $N_2$ does not because of a single communication error. If $N_1$ and $N_2$ receive $H$ after $t_1$ and $N_2$ does not receive $\sigma$ until $d_H$ (i.e. the deadline of $H$) due to bus overload, then at $d_H$, $N_1$ must deliver $S$ before $H$, and $N_2$ must deliver $H$ before $S$.

Due to the restriction mentioned above, this paper proposes an ordering algorithm, which establishes a total order for the messages of each class (i.e. hard and soft real-time) separately.

The paper is organized as follows: section 2 introduces some properties of the CAN bus which are essential to understand the approach. Section 3 presents the deadline-based timely message delivery scheme. Section 4 introduces the deadline-based total ordering scheme. Section 5 discusses how to reflect Lamport's precedence relation using the deadline-based total order. A summary concludes the paper.

2 Some Properties of CAN

The CAN-bus (BOSCH, 1991) is a priority bus targeted to operate in a noisy environment with speeds of up to 1 Mbit/s, exchanging small real-time control messages.

The key to understanding the CAN-Bus is the fact that all nodes scan the value of every bit while it is being transmitted. Hence, all correct nodes have a consistent view of every bit. CAN controllers are commonly connected by a wired AND circuit, i.e. whenever different bit values are sent by different nodes simultaneously, the logical AND function of the bit values is observed by all nodes (including the senders).

This consistent global view is exploited for a priority-based arbitration mechanism. According to this mechanism, as soon as the bus is idle, each node competing for the bus begins to send the arbitration field of its message, which mainly consists of an 11-bit or 29-bit identifier (depending on the "standard" or "extended" format - Figure 1). If at this time a node sends a '1' and senses
A) Standard Format

<table>
<thead>
<tr>
<th>11 bits identifier</th>
</tr>
</thead>
<tbody>
<tr>
<td>RTR-bit</td>
</tr>
<tr>
<td>IDE-bit (=‘0’)</td>
</tr>
</tbody>
</table>

B) Extended Format

<table>
<thead>
<tr>
<th>11 most significant bits of identifier</th>
<th>18 least significant bits of identifier</th>
</tr>
</thead>
<tbody>
<tr>
<td>SSR-bit (=‘1’)</td>
<td>RTR-bit</td>
</tr>
<tr>
<td>IDE-bit (=‘1’)</td>
<td></td>
</tr>
</tbody>
</table>

Fig. 1. The CAN arbitration field

a '0', it becomes aware of a collision, stops transmitting, and switches to the receiver mode. At the end of the arbitration field, only the node which is sending the message with the lowest arbitration field value, will be transmitting. Due to this arbitration technique, the identifier of a CAN-message serves as its priority, and the bus acts as a priority-based dispatcher.

The priority-based arbitration mechanism requires that different CAN nodes never simultaneously send messages with equal identifiers. This requirement must be satisfied by the communication software.

The CAN protocol provides efficient hardware-implemented error handling, which is based on various error detection mechanisms with a very high total coverage, and an approach to immediately signaling the error condition. These features of the CAN protocol ensure atomic broadcast delivery in most fault situations. However, if an error occurs within the transmission time of the two last bits of a data frame, some receivers might accept and other receivers reject the frame. Although the sender (or in case of immediate sender crash an alternative mechanism like Eager Diffusion (Rufino et al., 1998) or Shadow Retransmitter (Livani, 1998)) will retransmit the frame, the consistent ordering of the incoming messages in different sites is not trivial due to the possible transmission of other (high-priority) messages between the first and the second transmission of a frame. Another problem caused by such errors is the duplicate frame reception by some receivers.

In such situations, commercially available application level protocols for CAN, like CAL (CiA, 1993), SDS (Crovella, 1994), and DeviceNet (Noonen et al., 1994) manage to discard frame duplicates, but they fail to provide consistent message ordering among a group of receivers.

3 The Deadline-based Timely Message Transmission Mechanism

The global scheduling in a distributed system requires consensus between all participants about the use of shared system resources. The global schedule has to be enforced by all components, based on their local information. In
<table>
<thead>
<tr>
<th>8 bits</th>
<th>8 bits</th>
<th>13 bits</th>
</tr>
</thead>
<tbody>
<tr>
<td>Priority</td>
<td>TxNode</td>
<td>Message name</td>
</tr>
</tbody>
</table>

Fig. 2. Partitioning of a CAN-message identifier

a completely static system, a periodic global calendar is available and each component has its relevant entries referring to its activities in a global time scale. A global activity may only be started according to this schedule (Kopetz and Merker, 1985; Kopetz and Grünsteidl, 1994). In a more dynamic system where hard real-time, soft real-time, and non-real-time tasks coexist, things are more complicated. If a computing resource is free, a less critical (soft or non real-time) task may start computation and request resources. In this case, it must not lead to the timing failure of a hard real-time activity.

The scheduling of the network is a central part of the global resource scheduling problem. In the next section, a hybrid global scheduling approach for the CAN bus is described, which is appropriate for application systems with hard and soft real-time distributed activities. The hybrid scheduling mechanism is mainly based on a dynamic priority scheme, and resource reservation.

3.1 The dynamic priority scheme

The dynamic priority scheme of the hybrid bus scheduling algorithm implements the Least-Laxity-First Scheme. In order to realize LLF in a CAN network, the mapping of the transmission laxity into the message priority has to be defined, so that a message with a shorter transmission laxity wins the bus arbitration against a message with a longer laxity. In CAN the entire identifier is used as a priority field. However, the whole identifier cannot be used as a deadline-driven dynamic priority because of following reasons: Firstly, a subject-related message name should be encoded into the identifier in order to support subject-based addressing and exploit the message filtering features of the CAN controller hardware. Secondly, the sender node identifier must be included into the identifier field of CAN messages, to ensure that competing messages always have different identifiers. Note that this uniqueness cannot be achieved by subject-related unique naming of messages in a system with replicated objects. Figure 2 illustrates how the CAN identifier is partitioned in order to fulfill all requirements mentioned above.

In the priority field of a real-time message, the time remaining until its latest transmission starts (let's call it transmission laxity) is encoded. Choosing the LLF scheme is justified by the following facts: firstly, the local scheduling decision on each node can be simplified by a queue of outgoing messages sorted by the transmission laxity. Secondly, a laxity can easily be transformed into a unified scale of priorities. Thirdly, the global priority-based message
dispatching is performed by the arbitration mechanism of CAN at no cost. Due to the fact that the lower binary values of the priority field represent higher priorities, a shorter (lower) laxity is mapped to a lower value in the priority field.

Having a fixed value range \( \{P_{\text{min}}, \ldots, P_{\text{max}}\} \) for the priority field, a transmission laxity value \( \Delta L \) is mapped to a priority value \( P \), where \( P = \lfloor \Delta L / \Delta t_p \rfloor + P_{\text{min}} \) for \( \Delta L < (P_{\text{max}} - P_{\text{min}}) \cdot \Delta t_p \) and \( P = P_{\text{max}} \) for \( \Delta L \geq (P_{\text{max}} - P_{\text{min}}) \cdot \Delta t_p \). The modification of the dynamic priorities can be performed by periodically decreasing \( P \) once per \( \Delta t_p \).

The period \( \Delta t_p \) is called the priority slot. Since all laxity values \( \Delta L \geq (P_{\text{max}} - P_{\text{min}}) \cdot \Delta t_p \) are mapped to the same priority, the priority-based dispatcher (i.e. the CAN arbitration mechanism) cannot distinguish different laxity values which are greater or equal \((P_{\text{max}} - P_{\text{min}}) \cdot \Delta t_p \). Hence \((P_{\text{max}} - P_{\text{min}}) \cdot \Delta t_p \) is the time horizon of the proposed LLF-scheduler. The term time horizon denotes the amount of time in the future that can be correctly analyzed and planned by a time-based decision-maker, like an EDF or LLF scheduler.

The required time horizon of a scheduling mechanism depends on the timing requirements of the real-time application. In the following discussion, a CAN bus is assumed, where in worst case \( N \) transmitters attempt to send messages with equal laxity, and another one has a message with a larger laxity. If all laxity values are larger than the time horizon of the LLF scheduler, then they will be transformed into the same priority \( P_{\text{max}} \). Then if the message with the larger laxity wins the arbitration, and the remaining time is too short to schedule the other \( N \) messages after it, then the scheduler will fail because of too short a time horizon. In order to avoid a scheduling failure in this case, the time horizon should not be shorter than the time required for the transmission of an arbitrary set of \( N \) messages.

Assuming a maximum transmission time of \( \Delta T_{\text{max}} \), an inter-frame spacing of 3 bits, a maximum transmission failure rate of \( \lambda_{\text{max}} \), and a maximum time loss of \( \Delta T_{\text{fail}} \) per transmission failure, the time horizon \( \Delta H \) must satisfy the condition: \( \Delta H \geq N \cdot (\Delta T_{\text{max}} + 3) + [\Delta H \cdot \lambda_{\text{max}}] \cdot \Delta T_{\text{fail}} \). Thus, for the proposed LLF scheduler, a sufficient time horizon for correct LLF scheduling of messages is:

\[
\Delta H \geq \frac{N \cdot (\Delta T_{\text{max}} + 3)}{1 - \lambda_{\text{max}} \cdot \Delta T_{\text{fail}}} + \Delta T_{\text{fail}}
\]

Note that the above criterion is only a recommendation for the optimal LLF scheduling of soft real-time messages, the timely transmission of hard real-time messages is ensured by the dynamic TDMA scheme, as described below.
3.2 The dynamic TDMA scheme

In order to guarantee the timely transmission of hard real-time messages even in overload situations, all their occurrences have to be predicted and the respective transmission times have to be scheduled in advance in a calendar. Due to the highly critical nature of hard real-time messages, the scheduled transmission times must include worst-case error handling delays and retransmission times under anticipated fault conditions. A study of the worst-case delay times under different fault conditions is given in (Rufino and Verissimo, 1995).

3.2.1 The global TDMA calendar

The reserved time slots are entered into a calendar, which contains all resource reservations. The calendar is contained in each node of the system. This enables the system components to monitor each other’s temporal behavior. If a component claims reserved resources at the wrong time, other components will agree on its failure, and can initiate error recovery actions.

3.2.2 The impact of inaccurate global time

The scheduling approach for hard real-time communication requires access to a global time reference with bounded inaccuracy. Once a time slot is reserved, the respective action can be started locally. To guarantee that it does not interfere with another time slot, the time reference of all nodes must be synchronized. The lower the clock accuracy, the larger the minimum gap between two subsequent time slots in the global bus schedule (Figure 3). In order to provide a global time reference with high accuracy, a clock synchronization mechanism has to be applied, e.g. as described by (Gergeteit and Streich, 1994).
3.2.3 Exclusive access right in the reserved time-slot

Efficient handling of time slot reservations is supported by defining the end of the reserved time slot of a message \( m \) as its transmission deadline \( d_m \), and calculating its laxity as required by the underlying LLF scheme. The sender of a hard real-time message enforces its exclusive access rights during the reserved time slot by dynamically increasing the priority of the message according to its laxity. Due to this scheme, a hard real-time message gains the highest possible priority at its latest transmission time. In order to guarantee that different hard real-time messages are always assigned different priorities in the right order, their deadlines must at least differ by the length \( \Delta t_p \) of the priority slot (Figure 4).

Note that every message may be delayed by one other message, which may have started before it. \( \Delta T_{\text{block}} \) is defined as the longest possible blocking time due to the non-preemptive transmission of an arbitrary message. Since the transmission of a hard real-time message \( h \) (with a reserved time slot beginning at \( S_h \)) must start before \( S_h \), it must be ready until \( S_h - \Delta T_{\text{block}} \) (called the latest ready time \( LRT_h \)), and after \( LRT_h \) it must always win the arbitration process against all other messages.

From the previous discussion, following requirements are derived for the correct scheduling of hard real-time messages:

(R1) for each occurrence of a hard real-time message \( h \), an exclusive time-slot is reserved, which ends at its deadline \( d_h \);
(R2) the length \( \Delta R_h \) of the reserved time-slot of a hard real-time message \( h \) is greater or equal to the worst-case transmission time of the message, including all error handling delays and retransmission times under anticipated fault conditions;
(R3) every hard real-time message \( h \) is ready for transmission until its latest ready time \( LRT_h = S_h - \Delta T_{\text{block}} \);
(R4) during its critical interval \([LRT_h \ldots LST_h]\) (Figure 5), a hard real-time
Fig. 5. The critical interval of a hard real-time message
message \( h \) wins the bus arbitration against every other message in the system.

**Claim 2** If the requirements R1 through R4 are fulfilled in a CAN bus, then every hard real-time message will be transmitted timely under anticipated fault conditions.

**PROOF.** Assume that a hard real-time message \( h \) occurs with the deadline \( d_h \). Due to R1 and R2 a time-slot \([d_h - \Delta R_h \cdots d_h]\) is reserved for \( h \). Due to R3, \( h \) is ready for transmission until \( d_h - \Delta R_h - \Delta T_{\text{block}} \). If \( h \) is not successfully transmitted until \( LRT_h \), then it participates at least in a bus arbitration at \( t\in[LRT_h\cdots S_h] \), and due to R4 \( h \) wins the arbitration at \( t \). Due to R2 the time loss caused by successive communication failures is not more than \( \Delta R_h - \Delta T_h \). Hence at least one correct message transmission is started in the interval \( [t\cdots t + \Delta R_h - \Delta T_h] \subseteq [LRT_h\cdots LST_h] \), and due to R4, \( m \) wins the bus for this correct transmission, and finishes until \( t + \Delta R_h \leq D_m \). □

The requirements R1, and R2 are ensured by a calendar-based global schedule, which is distributed to all nodes, which also have access to synchronized clocks. In order to reserve the right amount of time for a message in the calendar, the time-slot length must be calculated based on the message length and the fault hypothesis. For a message \( h \) with \( b_h \) bytes of data, the maximum length of the message including header and bit-stuffing is \( L_h = 75 + [b_h \times 9.6] \). Under the assumption of \( f \) consecutive transmission failures due to bus/controller errors, the required minimum time-slot length is: \( \Delta R_h = (L_h+18)\times f + L_h + 3 \) bit-times. The requirement R3 must be satisfied by the synchronization of the local task schedules with the global communication schedule.

To satisfy the requirement R4, the algorithm uses the following rules to ensure that during the critical interval between the latest ready time and the latest transmission start time of a hard real-time message \( h \) (Figure 5), all other messages in the system have lower priorities than \( h \).
The minimum priority field value for non-real-time messages is higher than the maximum for the real-time messages, i.e. \( P_{\text{min}}^{\text{NRT}} > \max\{P_{\text{max}}^{\text{SRT}}, P_{\text{max}}^{\text{HRT}}\} \). Thus, real-time messages always have higher priorities than non-real-time messages.

The minimum priority field value for soft real-time messages is higher than the maximum of any hard real-time message at the latest ready time, i.e.

\[
P_{\text{min}}^{\text{SRT}} > P_{\text{min}}^{\text{HRT}} + \left[ \frac{\Delta R_{\text{max}} + \Delta T_{\text{block}}}{\Delta t_p} \right]
\]

Thus, after its latest ready time, a hard real-time message has a higher priority than any soft real-time message.

If the transmission of a hard real-time message is not started until its latest transmission start time, the transmission request is cancelled.

The priority slot \( \Delta t_p \) must not be longer than the minimum distance between the ends of any two consecutive reserved time-slots in the system, minus \( \Delta G_{\text{min}} \).

The rule r3 ensures that the transmission of a hard real-time message is either terminated or cancelled until the beginning of the reserved time-slot of the next scheduled hard real-time message, so it does not occupy reserved resources of the next hard real-time message. The rule r4 ensures that – within its critical interval – a hard real-time message has a higher priority than the next scheduled hard real-time message. Thus, within a reserved time-slot, there is no conflict between different hard real-time messages.

3.2.4 Calculation of the dynamic priorities.

According to the rule r1, the total available priority range must be split into two separate sub-ranges for real-time and non real-time communication. In the current design, the priority field consists of 8 bits, resulting in a range from zero to 255. The priority range \( \{0 \ldots 191\} \) is assigned to the non-real-time communication, i.e. \( P_{\text{min}}^{\text{NRT}} = 192 \) and \( P_{\text{max}}^{\text{NRT}} = 255 \).

The priority range \( \{0 \ldots 191\} \) is fully used for the hard real-time communication, i.e \( P_{\text{min}}^{\text{HRT}} = 0 \) and \( P_{\text{max}}^{\text{HRT}} = 191 \). Also, the largest priority value for soft real-time messages is set to \( P_{\text{max}}^{\text{SRT}} = 191 \). The value \( P_{\text{min}}^{\text{SRT}} \) is calculated according to the rule r2 as:

\[
P_{\text{min}}^{\text{SRT}} = P_{\text{min}}^{\text{HRT}} + \left[ \frac{\Delta R_{\text{max}} + \Delta T_{\text{block}}}{\Delta t_p} \right] + 1 = \left[ \frac{\Delta R_{\text{max}} + \Delta T_{\text{block}}}{\Delta t_p} \right] + 1
\]

For example, given a single fault tolerant system, \( \Delta R_{\text{max}} \geq 323 \Delta t \) is mandatory. Assuming a \( \Delta t_p = 200 \Delta t \), and \( \Delta T_{\text{block}} = 154 \Delta t \), the value of \( P_{\text{min}}^{\text{SRT}} \)
will be 3.

Non real-time messages are assigned fixed priorities \( P_N \in \{ P_{\text{min}}^{\text{NRT}} \ldots P_{\text{max}}^{\text{NRT}} \} \). The priority of a hard real-time message with the laxity \( \Delta L \) is calculated as:

\[
P_H(\Delta L) = \min \left\{ P_{\text{min}}^{\text{HRT}} + \left\lfloor \frac{\Delta L}{\Delta t_p} \right\rfloor, P_{\text{max}}^{\text{HRT}} \right\} = \min \left\{ \left\lfloor \frac{\Delta L}{\Delta t_p} \right\rfloor, 191 \right\}
\]

Soft real-time messages must compete against hard real-time messages according to the LLF scheme, until their priority becomes equal to \( P_{\text{min}}^{\text{SRT}} \). After that, the priority of a soft real-time message remains constant. Thus the priority of a soft real-time message with the laxity \( \Delta L \) is calculated as:

\[
P_S(\Delta L) = \max \left\{ P_{\text{min}}^{\text{SRT}}, P_H(\Delta L) \right\}
\]

The assignment of the dynamic priorities is efficiently implemented as follows. The hard and soft real-time messages are inserted into two separate queues, each sorted by deadline. Each time that a new message moves to the head of a queue, its priority is calculated based on its current laxity. Once the priority of a message is assigned, it is updated by periodic decrementation at each \( \Delta t_p \).

### 3.3 Schedulability of hard real-time messages

Since the dynamic TDMA scheme guarantees the timely transmission of every hard real-time message by reserving a time-slot, the number of the schedulable hard real-time messages depends on the length of the time-slots and the length of the gaps between time-slots.

In a system tolerating \( f \) consecutive transmission failures, the required length of the reserved time-slot of a hard real-time message \( h \) can be calculated as:

\[
\Delta R_h = \Delta T_h + 3 + f \times \Delta T_{\text{fail}}
\]

As an example, in a single-fault tolerant system, a time-slot of \( 323 \times \Delta bt \) is required for arbitrary messages. The minimum gap between consecutive time-slots depends on the accuracy of the clock synchronization mechanism in the system. In a CAN bus system, a maximum clock inaccuracy of \( \pm 20 \mu s \) can be achieved, e.g. by the algorithm given in (Gergeleit and Streich, 1994). Assuming a maximum clock inaccuracy of \( \pm 20 \mu s \), the minimum gap \( \Delta G_{\text{min}} \) between two consecutive time-slots must be 40\( \mu s \). Under these assumptions, a time-slot can be planned each 363\( \mu s \). In this case, up to 2754 hard real-time messages per second could be guaranteed.
A sufficient condition for the scheduling of hard real-time messages is that for each message an appropriate time-slot can be inserted into the calendar schedule with the required frequency, with enough of a gap between each pair of consecutive time-slots. This, of course, depends on the application parameters, e.g. individual message periods, etc.

4 The Deadline-Based Total Ordering Scheme

Based on a feasible and flexible real-time scheduling policy as described in the previous section, combined with a reliable broadcast delivery mechanism (Livani, 1998; Rufino et al., 1998), following assumptions can be made about the communication system in a CAN bus:

(A1) Hard real-time messages are transmitted timely (i.e. until their transmission deadline) under anticipated fault scenarios, even in overload situations.

(A2) Soft real-time messages are scheduled by static or dynamic priorities, but their deadlines may be missed in overload situations.

(A3) If a non-faulty receiver receives a message, then eventually all non-faulty receivers receive the message.

This section introduces a scheme to achieve globally consistent ordering of multicast messages in a CAN-based system. The algorithm presented here relies on the assumptions A1 through A3 in order to achieve atomic multicast with minimum communication overhead, based on the knowledge about the message transmission deadlines. The transmission deadline of a message denotes the time, at which the message must be successfully transmitted to all non-faulty destination sites. Since the transmission deadline is tightly related to the time, where the sender expects the message delivery to all receiving application objects, this ordering mechanism allows an application specific ordering, which is related to temporal requirements. Although this approach is similar to some other ones known from much previous work (Cristian et al., 1985; Schneider, 1990), it considers the late transmission of soft real-time messages in overload situations. Another advantage of this scheme is that it needs no additional communication to establish a globally consistent ordering decision.

Let \( m \) and \( m' \) be two real-time messages with transmission deadlines \( d_m \) and \( d_{m'} \). Then the deadline-based ordering algorithm implements the following rule:

(O1) \( d_m < d_{m'} \Rightarrow del_j(m) \rightarrow del_j(m') \)
This means that if the deadline of the message \( m \) is before the deadline of the message \( m' \), then \( m \) must be delivered to destination objects before \( m' \). If the underlying protocol layer adjusts the transmission deadlines of hard real-time messages to reserved time-slots (cf. section 3), then the delivery order of hard real-time messages can be always established by this rule.

However, when the deadlines of different messages are equal, the following rule must be used to ensure the globally consistent decision on delivery order of messages.

\[
\text{O2)} \quad d_m = d_{m'} \Rightarrow (del_j(m) \rightarrow del_j(m') \Leftrightarrow header_m < header_{m'})
\]

Where \( header_m \) consists of a tuple \( (d_m, \text{sender}_m, \text{subject}_m) \) and identifies a message uniquely. This rule requires that different messages sent by the same sender with the same subject have different transmission deadlines, otherwise an additional sequence number (or "toggle-bit") must be provided in the header. Note that this requirement is also necessary for distinguishing duplicate frames from successive frames.

### 4.1 Protocol Termination and End-to-End Delivery

An atomic multicast protocol may terminate and deliver a message to destination objects, if and only if a) the message is received and accepted by all non-faulty destination sites, and b) the message can be consistently ordered, which means that no other message will arrive later, which must precede the current message according to the ordering criteria.

a) **Consistent Message Reception in CAN**

In order to minimize the communication overhead, the nodes must not exchange explicit information about the status of their message queues. Thus they have to conclude this information from implicit knowledge. Since the underlying message transfer protocol guarantees the timely transmission of hard real-time messages under anticipated fault conditions, the protocol can assume that a hard real-time message is consistently transmitted to all receivers at its deadline. Hence a receiver can deliver the message to the destination objects at the transmission deadline.

For soft real-time messages, however, the transmission deadline may be missed because of bus overload, and hence, late transmission of the message is still possible. In these situations, the deadline cannot be used to specify the time when the message may be delivered. However, in such situations the following considerations lead to a decision criterion:
Claim 3 If a node has received a message $m$, and then another message with lower priority is observed on the CAN bus, or the bus is idle, then the sender of $m$ will not retransmit it in future.

PROOF. Assume that a message $m$ is transmitted at least once on the CAN bus. Further, assume that the sending CAN controller still attempts to retransmit $m$ due to the inconsistent transmission. According to the CAN specification (BOSCH, 1991), the sender will try to retransmit $m$ "immediately", thus no bus idle period will be observed before the retransmission of $m$. Furthermore, no lower-priority message will be transmitted before the retransmission of $m$, because $m$ will win the arbitration process against any lower-priority message.

As a consequence, the receivers of a soft real-time message $m$ can be sure that either the sender is crashed, or the transmission was successful, as soon as they detect either a bus idle period or a lower-priority message after receiving $m$.

If a frame is accepted by a subset of the receivers, and rejected by other receivers, the immediate sender crash may lead to inconsistent message reception. This failure must be tolerated by having some nodes retransmit the frame. The Eager Diffusion Protocol (Rufino et al., 1998) retransmits every frame at least by one receiver. However, due to processing delay the Eager Diffusion Protocol cannot guarantee the immediate retransmission. The ordering scheme presented in this paper relies on dedicated Shadow Retransmitters (Livani, 1998), which guarantee immediate retransmission whenever an inconsistent message reception is possible. The Shadow retransmitters also transmit an ultra-low priority frame (called trailer) at the end of each non-broken sequence of CAN frames. So an idle bus can be detected by observing a trailer on the bus.

From the previous discussion, following properties are derived:

(P1) Any receiver of a hard real-time message $m$ with deadline $d_m$, can assume that $m$ will be received by all sites until $d_m$.

(P2) Any receiver of a soft real-time message $m$ can assume that the message has been received by all sites, if it observes a frame with a lower priority on the bus after receiving $m$.

b) Achieving a Consistent Message Order in CAN

In case of hard real-time messages, the knowledge of time is sufficient for stabilizing the ordering decision: after a deadline $d_m$, no hard real-time message with an earlier deadline will arrive. But this statement is not true in case of soft
real-time messages. Here, the following assumption constitutes the constraint necessary to find a stable ordering decision criterion:

(A4) Every real-time message \( m \) is ready to transmit before \( d_m = \Delta T_{\text{max}} \), with \( \Delta T_{\text{max}} \) being the maximum time required for a single transmission of an arbitrary frame.

The assumption (A4) concludes the following:

**Claim 4** Assume a deadline-based message priority scheme (e.g. as described in section 3), where a soft real-time message with a higher priority has an earlier deadline than a soft real-time message with a lower priority. If after the transmission deadline \( d_m \) of a soft real-time message \( m \) another message with lower priority is transmitted on the CAN bus, or the bus is idle, then no other soft real-time message \( m' \) with a deadline \( d_{m'} \leq d_m \), will be transmitted later.

**Proof.** Assume that a message \( m' \) with deadline \( d_{m'} \leq d_m \) is pending for transmission at the time \( t > d_m \). Because of the deadline-based priority assignment the priority of \( m' \) is not lower than the priority of \( m \). Due to (A4), \( m' \) has been pending for transmission at least since \( d_{m'} = \Delta T_{\text{max}} \), hence at least since \( d_m - \Delta T_{\text{max}} \). Hence the sender of \( m' \) must have been trying to transmit \( m' \) at the beginning of every bus-idle period since \( d_m - \Delta T_{\text{max}} \). Therefore no idle bus can be observed between \( d_m \) and the successful transmission of \( m' \). Also, no lower-priority message can be transmitted on the bus before \( m' \), because \( m' \) wins the arbitration process against any lower-priority message. Thus, if after \( d_m \) a message with a lower priority than \( m \) is transmitted on the CAN bus, or the bus is idle, then no message \( m' \) with \( d_{m'} \leq d_m \) will be transmitted later. 

From the previous discussion, following property is derived:

(P3) For any soft real-time message \( m \), no preceding soft real-time message will arrive later, if after the transmission deadline \( d_m \) a lower-priority message is observed on the bus.

4.2 Atomic Multicast Delivery Algorithm

The proposed atomic multicast delivery algorithm is based on the following rules:
Order: Messages are ordered by their deadlines. In case of equal deadlines, the value of the message header is used for the order decision.

Time1: Every received hard real-time message is delivered at its deadline. (This realizes the deadline-based order implicitly)

Time2: Every received soft real-time message must be delivered as soon as either another message with later deadline, or a bus idle time is observed after its transmission deadline.

Due to the rule (Time2), the delivery of a soft real-time message \( m \) may be delayed until \( d_m + \Delta T_{\text{max}} \). This must be considered when calculating the transmission deadline of any soft real-time message.

In the following, the atomic multicast delivery algorithm is presented. The algorithm assumes the availability of a real-time clock, which is synchronized globally with a bounded inaccuracy.

\texttt{Atomic\_multicast\_delivery}

\texttt{initialization:}
\begin{verbatim}
SRT\_q \leftarrow \text{empty}\_q;
HRT\_q \leftarrow \text{empty}\_q;
nextHRT\_delivery \leftarrow \text{eternity};
\end{verbatim}

\texttt{receive(m,t): /* \( m \) is sent or received at \( t \) */}
\begin{verbatim}
if m.dest\_group \in my\_groups \textbf{and} m.category = HRT then
    HRT\_q.insert (m);
    if m.dl < nextHRT\_delivery then
        nextHRT\_delivery \leftarrow m.dl;
        set\_wakeup (nextHRT\_delivery);
else
    if m.dl \leq t then
        /* overload! Deliver preceding SRTM */
        while SRT\_q.head.dl < m.dl or SRT\_q.head < m do
            mx \leftarrow SRT\_q.gethead;
            SRT\_deliver (mx, mx.dest\_group);
    else
        /* Deliver SRTM with passed DL */
        while SRT\_q.head.dl < t do
            mx \leftarrow SRT\_q.gethead;
            SRT\_deliver (mx, mx.dest\_group);
end receive;
\end{verbatim}

\texttt{wakeup(): /* invoked when the wake-up time is reached */}
\begin{verbatim}
mx \leftarrow HRT\_q.gethead;
HRT\_deliver (mx, mx.dest\_group);
if HRT\_q \neq \text{empty}\_q then
\end{verbatim}

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5 Global Ordering of Events

The deadline-based ordering of multicast messages enables a consistent global order between different events observed in the distributed system. In this paper an event is called a global event, if the object which observes it, sends an atomic multicast to enforce a global observation of the event by all concerning objects. As seen by application objects, the order of global events is the same as the globally consistent delivery order of their notification messages. This globally consistent delivery order of messages is established by the atomic multicast protocol described in section 3.2. Note that if a set of events has to be globally ordered, then all multicast messages propagating those events must belong to the same class (hard or soft real-time).

5.1 Achieving Causal Ordering of Events

If global events have to be ordered according to their causality, following cases can be observed:

A: Let $e$ and $e'$ be two global events locally observed by an object $i$, with a causal precedence relation $e \rightarrow e'$, and let $m$ and $m'$ be messages, which propagate the events $e$ and $e'$ in the system. Let $d_m$ denote the deadline of $m$. In order to globally agree on the order $e \rightarrow e'$, for every object $j$, the relation $\text{del}_j(m) \rightarrow \text{del}_j(m')$ is sufficient. Hence, due to the rule (O1), the relation $d_m < d_{m'}$ is sufficient. This means, that in order to reflect a precedence order $e \rightarrow e'$ globally, $i$ must assign deadlines $d_m < d_{m'}$ to the messages propagating $e$ and $e'$ in the system.

B: Let $e$ and $e'$ be two events observed by two different objects $i$ and $j$, and
$e \rightarrow e'$, then (according to Lamport’s definition of precedence relation (Lamport, 1978), and the discussion in section 3.1) following conditions hold:

1. Event $e$ is observed by $i$ at local time $T^i(e)$.
2. Event $e'$ is observed by $j$ at local time $T^j(e')$.
3. There is a message $m$ sent by $i$ at the local time $T^i(send_i(m))$, which propagates the event $e$, where $T^i(e) \leq T^i(send_i(m)) < d_m$.
4. There is a message $m'$ sent by $j$ at the local time $T^j(send_j(m'))$, which propagates the event $e'$, where $T^j(e') \leq T^j(send_j(m')) < d_{m'}$.
5. If $j$ receives $m$, then it receives $m$ at the local time $T^j(del_j(m))$, where $d_m \leq T^j(del_j(m))$, and $T^j(del_j(m)) \leq T^j(e')$. Thus due to (4) it follows that $d_m < d_{m'}$.
6. If $j$ does not receive $m$, then there is a set of messages $m_1, m_2, m_3, \ldots, m_n$, and a set of objects $k_1, k_2, k_3, \ldots, k_n$, where firstly, $k_1$ receives $m$ and sends $m_1$, $k_2$ receives $m_1$ and sends $m_2$, \ldots, $k_n$ receives $m_{n-1}$ and sends $m_n$, and $j$ receives $m_n$. Secondly, $del_{k_1}(m) \rightarrow send_{k_1}(m_1), del_{k_2}(m_1) \rightarrow send_{k_2}(m_2), \ldots$, and $del_{k_n}(m_{n-1}) \rightarrow send_{k_n}(m_n)$.

Thirdly, $T^j(del_j(m_n)) \leq T^j(e')$. In this case, $d_m \leq T^{k_1}(del_{k_1}(m)) < T^{k_1}(send_{k_1}(m_1)) < d_m \leq T^{k_2}(del_{k_2}(m_1)) < T^{k_2}(send_{k_2}(m_2)) < d_m \ldots < d_m \leq T^{k_n}(del_{k_n}(m_{n-1})) \leq T^j(e')$. Again, due to (4) it follows that $d_m < d_{m'}$.

In either case (i.e. A; B–5; B–6), every object $l$ in the system which receives $m$ and $m'$, will establish the delivery order $del_l(m) \rightarrow del_l(m')$ due to the relation $d_m < d_{m'}$. Consequently, $l$ will establish the precedence order $e \rightarrow e'$. Thus, if $e$ and $e'$ are two global events observed in the distributed system, and $e \rightarrow e'$ according to Lamport’s definition of precedence relation, then all non-faulty objects in the system – which are addressed by messages propagating $e$ and $e'$ – will be notified about $e$ before $e'$, and hence they will establish consistently the precedence order $e \rightarrow e'$.

6 Conclusion

The paper introduced a real-time multicast ordering scheme, which achieves consistent delivery ordering of multicast messages using the message transmission deadlines.

The protocol termination time depends on the message class. The protocol relies on a medium access protocol, which guarantees timely and reliable transmission of hard real-time messages to all destination nodes (Livani, 1998). Hence, for a hard real-time message the protocol terminates at the transmission deadline and delivers the message to destination objects timely. Under ‘normal’ load conditions, the protocol delivers soft real-time messages up to
one frame transmission time later than their transmission deadlines (cf. rule Time2 in section 3.2). However, in overload situations, the protocol may delay the delivery of soft real-time messages until the end of the overload period plus one frame transmission time.

The mechanism ensures consistent multicast ordering in presence of communication failures leading to inconsistent global view of the CAN bus status among non-faulty sites. Common high-level communication protocols designed for CAN, do not provide consistent multicast delivery order among all application objects in these failure situations.

It was shown, that causal order between events can be globally established by using the deadline-based ordering approach.

An important benefit of this scheme is its efficiency: the algorithm requires no additional communication, like acknowledgements etc. In the Controller Area Network, communication bandwidth is a serious bottle-neck. While the computing power of hardware nodes is increasing rapidly, the bandwidth of CAN will remain limited to a maximum of 1 Mbits/sec. This was the main motivation to develop a multicast scheme which minimizes communication amount for the price of some computational overhead.

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