

R-MAC: An Energy-Efficient MAC Protocol for Underwater Sensor Networks *

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Abstract

Underwater sensor networks are significantly different from terrestrial sensor networks in that sound is mainly used as the communication medium. The long propagation delay and limited bandwidth of acoustic channels make the existing MAC protocols designed for radio networks either unpractical or not energy efficient for underwater sensor networks. In this paper, we propose a reservation-based MAC protocol, called R-MAC. The major design goals of R-MAC are energy efficiency and fairness. R-MAC schedules the transmissions of control packets and data packets to avoid data packet collision completely. The scheduling algorithms not only save energy but also solve the exposed terminal problem inherited in RTS/CTS-based protocols. Furthermore, the scheduling algorithms allow nodes in the network to select their own schedules, thus loosening the synchronization requirement the protocol. Additionally, R-MAC supports fairness. By simulations, we show that R-MAC is an energy efficient and fair MAC solution for underwater sensor networks.

1 Introduction

Underwater sensor network has emerged as a powerful technique for aquatic applications, and it has attracted more and more attention from the networking research community recently [16, 11, 2, 7, 5].

Due to the dense network deployment and the shared communication medium, an efficient medium access control (MAC) protocol is very important to the final performance of underwater sensor networks. Different applications have different requirement on MAC protocols. In this paper, we aim to design a MAC protocol for long term aquatic monitoring applications [5]. This type of applications are not sensitive to the end-to-end delay

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and generate sporadic traffic unevenly distributed spatially and temporally. For such applications, sensor nodes are usually deployed densely, with tens to hundreds of meters apart. The most important goal of the MAC design for such underwater sensor networks is to *resolve data packet collision efficiently in terms of energy consumption*. This is because sensor nodes in underwater sensor networks are usually powered by batteries, and it is difficult to change or recharge these batteries in harsh underwater environments. *Fairness* is another goal of our MAC protocol, as it is very important for in-network data processing. Since data are usually generated by the neighboring sensor nodes that are highly temporally related, it is desirable to deliver temporal related data to some node at the same time for in-network processing. Other properties such as end-to-end latency, throughput and channel utilization are desirable but not hard requirements in our design.

In short, for our targeted applications, the design goals of our MAC protocol is to *resolve data packet collision and support fairness in an energy efficient way*.

There are numerous MAC protocols for designed radio networks. However, we can not directly adapt these protocols to underwater sensor networks due to the significant difference between underwater sensor networks and terrestrial (radio) sensor networks. In underwater sensor networks, acoustic channel is used as the communication method. The propagation speed of sound in water is about 1500 m/s, which is 5 orders of magnitude lower than that of radio. The low propagation speed results in a high propagation delay even for communication between two neighbors. Moreover, the available bandwidth of acoustic channels is typically less than 15 KHz, which is much narrower compared with that of RF channels. These unique acoustic communication characteristics pose many challenges when applying existing protocols in underwater sensor networks.

In general, MAC protocols can be roughly divided into two categories: contention-free protocols and contention-based protocols. Contention-free protocols include TDMA, FDMA and CDMA, where communication channels are separated in time, frequency or code domains. It is common wisdom that FDMA is unsuitable for underwater sensor networks because of the narrow available bandwidth. There are some researches on TDMA and CDMA for underwater networks. However, some problems inherent in these methods have not been well addressed in acoustic networks. For example, the synchronization problem in TDMA and near-far problem in CDMA. Thus, the feasibility of these protocols in underwater sensor networks is unclear.

Contention-based protocols includes random access methods [1, 12, 13] and collision avoidance methods [8, 4, 6]. In a random access protocol, the sender sends packets without coordination. Thus packet avoidance is totally probabilistic. While in a collision avoidance protocol, the sender and receiver capture the medium through control packet exchange before data transmission.

There are many collision avoidance protocols, among which RTS/CTS-based protocols are widely used. The

performance of random access methods and RTS/CTS-based approaches in underwater sensor networks is determined by many factors. [17] has shown that RTS/CTS-based methods outperform random access approaches (in terms of energy efficiency) in a network with short transmission ranges, dense neighborhood and bursty traffic. These properties are close to those in our targeted underwater sensor networks. Thus, it appears promising for us to explore the RTS/CTS based approaches.

However, the long propagation delay in underwater sensor networks makes it too energy consuming to use RTS/CTS to eliminate the data packet collisions completely. As shown in [6], in order to completely avoid data packet collision, two conditions have to be satisfied: 1) the duration of RTS should be greater than the maximum propagation delay; and 2) the duration of CTS should be greater than that of RTS plus twice the maximum propagation delay plus the hardware transmit-to-receive transmission time. It is easy to satisfy these two conditions in terrestrial radio sensor networks. However, it is too expensive to do so in underwater sensor networks where the maximum propagation delay is usually very long, as makes the size of control packets unacceptably large.

If we relax the requirement of completely avoiding data packet collisions, we can use a modified RTS/CTS-based method as in [17]. Although the modified method can not avoid data packet collisions completely, it can reduce the data packet collisions to a low level. However, the modified RTS/CTS method is not energy efficient. In order to avoid packet collisions, a node has to listen the channel during the active time period. On the other hand, the high propagation delay prolongs the active time period. Therefore, the high propagation delay results in more energy waste on overhearing and idle state when RTS/CTS exchange is used in underwater sensor networks.

In this paper, we concentrate on the collision avoidance approaches for underwater sensor networks. We propose a reservation-based MAC protocol, called R-MAC, to achieve the objectives of energy efficiency and fairness. In R-MAC, we do not use the RTS/CTS message exchange to avoid data packet collisions. Instead, we schedule the transmission of control packets and data packets at both the sender and the receiver to avoid data packet collisions completely. The scheduling algorithms address the problem caused by the high propagation delay efficiently. Moreover, R-MAC supports fairness. Additionally, we adopt a new ARQ technique, burst-based acknowledgment, in R-MAC to improve channel utilization. The burst-based acknowledgment combined with the scheduling algorithms solves the exposed terminal problem and improves the network throughput.

The rest of the paper is organized as follows. In Section 2, we present R-MAC and the scheduling algorithms in details. In Section 4, we evaluate the performance of R-MAC using simulations. In Section 5, we briefly discuss some related work. In Section 6, we conclude our work and discuss some future directions.

2 R-MAC Protocol Design

In this section, we first brief the basic ideas of R-MAC, then we describe the three phases of R-MAC in details. After that, we focus on the scheduling algorithms on both sender and receiver. Finally, we discuss several critical issues in R-MAC design.

2.1 Overview of R-MAC

In R-MAC, to reduce the energy waste on idle state and overhearing, each node works in listen and sleep modes *periodically*. The durations for listen and sleep are the same for all nodes, and each node *randomly* selects its own schedule, as means that no centralized scheduling and synchronization are required in R-MAC. For any node, if there is no traffic in its neighborhood, it simply listens and sleeps periodically. When a node (i.e., sender) wants to send data to another node (i.e., receiver), R-MAC employs a reservation-based approach to synchronize, in a distributed way, transmissions to avoid data collisions.

R-MAC has three phases, namely, *latency detection*, *period announcement*, and *periodic operation*. The first two phases are used to synchronize nodes in the neighborhood and the third one is for listen/sleep operations. A node in the latency detection phase detects the propagation latency to all its neighbors. In the period announcement phase, each node randomly selects its own listen/sleep schedule and broadcasts this schedule. The data (if there are any) are transmitted in the periodic operation phase. Next we discuss the three phases in details.

2.2 Phase One: Latency Detection

In this phase, all nodes power on. Each node randomly selects a time to broadcast a control packet, called *Neighbor Discovery packet*, denoted as *ND*. Upon receiving NDs from its neighbors, a node records the arrival times of these NDs, then randomly selects a time to transmit an acknowledgment packet, denoted as *ACK-ND*, which has the same packet size as ND, for each of the NDs it receives. In each ACK-ND, the node specifies the duration from the arrival time of the ND packet to the transmission time of this ACK-ND packet, I_2 . After receiving an ACK-ND, a node computes the interval from the time that the corresponding ND packet is transmitted to the arrival time of the ACK-ND, I_1 . Then the propagation latency, L , between the two nodes can be calculated as $L = \frac{I_1 - I_2}{2}$. Therefore, the propagation latency L between two nodes is the interval from the time the first node sends the first bit of a packet to the time the second node receives the last bit of the packet.

An example is illustrated in Figure 1. Node *A* sends an ND packet and records the time. Upon receiving this packet, node *B* randomly delays some time period I_B and sends an ACK-ND packet back to node *A*. Node *B* specifies in the ACK-ND packet the time interval I_B , its ID and the ID of ND packet. Upon receiving this packet,

node A computes the time interval from the transmission time of its ND packet to the arrival time of ACK-ND packet, I_A . Then node A calculates the propagation latency to node B as $L_{AB} = \frac{I_A - I_B}{2}$.

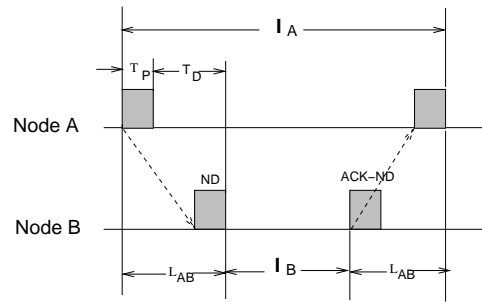


Figure 1. Latency measurement

Propagation latency L includes transmission delay and propagation delay. Since the system and network architecture in sensor nodes are relatively simple, the transmission delay is mainly determined by the hardware and the size of the packet. Thus its variance is negligible. The propagation delay is mainly depends on the sound speed in water, which might be affected by many factors such as temperature and pressure. However, for a short time period, it is reasonable to assume that the sound speed in water is constant, i.e., propagation delay does not change in a short time period. Therefore, propagation latency L is accurate and stable for a short time period. With time going, latency measurements become more and more inaccurate due to clock drift and varying sound speed. Thus, it is desirable that these measurement will be updated after a period of time, as can be done through the message exchange in the third phase.

After the latency detection phase, each node records the propagation latencies to all of its neighbors.

2.3 Phase Two: Period Announcement

In this phase, each node randomly selects its own start time of the listen/sleep periodic operations (i.e., the third phase) and broadcasts this time (we also call it *schedule*). After receiving broadcast packets, each node converts the received times (schedules) to its own time (schedule).

As shown in Figure 2, node A , randomly selects its listen/sleep schedule, and announces this schedule by broadcasting a synchronization packet, denoted as *SYN*. There are two fields in this packet: node A 's ID and time interval I_A , which specifies the interval from the time to send *SYN* to the beginning time of its third phase. Upon receiving a *SYN* packet from node A , node B calculates the time interval from the arrival time of this *SYN* packet to the starting time of its third phase, I_B . Then node B converts the schedule of node A relative to its own schedule by $I_B - I_A + L_{AB}$, where L_{AB} is the propagation latency from node A to node B .

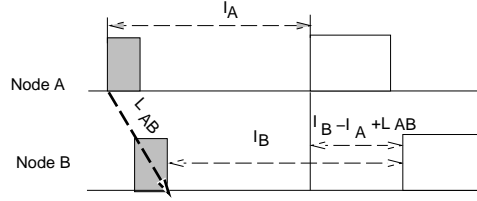


Figure 2. Schedule conversion

In the second phase of R-MAC, each node records the schedules of its neighbors relative to its own schedule. In the real implementation, the first two phases can run multiple rounds to make sure that all the nodes in the network have complete information about their neighbors.

2.4 Phase Three: Periodic Operation

In this phase, nodes wake up and sleep periodically. We call one listen/sleep cycle as one *period*. All nodes have the same periods. One period, denoted as T_0 , consists of two parts: listen window, denoted as T_L , and sleep window T_S . Thus $T_0 = T_L + T_S$.

In R-MAC, nodes communicate through *REV/ACK-REV/DATA/ACK-DATA* message exchange, where *REV* denotes the reservation packet, *ACK-REV* is the acknowledgment packet for *REV*, and *ACK-DATA* is the acknowledgment packet for data *DATA*. All the control packets in R-MAC, namely *REV*, *ACK-REV* and *ACK-DATA* have the same size, which is much smaller than that of data packets. When a node has data to send, it first sends a *REV* to reserve a time slot at the receiver. If the receiver is ready for data transmission, it will notify all its neighbors about the reserved time slot by *ACK-REVs*. Upon receiving *ACK-REVs*, all the nodes other than the sender keep silent in their corresponding time slots, and the sender can send data at the reserved time slot. In R-MAC, data are transmitted in a burst. A node queues its data for the same receiver until it captures the channel, then injects all the queued data. The receiver sends back an *ACK-DATA* to the sender at the end of the burst transmission. In other words, in order to reduce the control packet overhead and improve the channel utilization, the receiver acknowledges per burst instead of per packet. We refer to this technique as *burst-based acknowledgement* in this paper.

R-MAC treats *ACK-REVs* as the highest priority packets and reserves the first part of the listen window exclusively for *ACK-REV* packets. We call this reserved part *R-window*, denoted as T_R , which is the maximum possible duration of a control packet. The rationale of this design is the following: *ACK-REV* is used by the receiver to notify its neighbors of not interrupting the subsequent data transmission. If a node misses an *ACK-REV*, it possibly interferes the subsequent data transmission. As for the case of missing a *REV* or *ACK-DATA*,

no data collisions will be caused. In R-MAC, nodes only have to sense the channel in their R-windows to get the information about the subsequent data transmission. If a node receives an ACK-REV in its R-window, then this node knows the duration of the subsequent data transmission and keeps silent during that time period. However, when a node senses the channel busy in its R-window, but can not receive an ACK-REV clearly (i.e., there is a ACK-REV collision), it will back off. When a node is in backoff state, it still needs to sense the channel in its R-window and updates the usage information of the channel in its neighborhood.

Since R-windows are designed for receiving ACK-REVs, all other types of packets (including REV, DATA, ACK-DATA) have to avoid R-windows. To achieve this purpose, in R-MAC, all nodes have to carefully schedule the transmission of control and data packets. The scheduling algorithms at both the sender and receiver should guarantee that only ACK-REVs can propagate to any node in its R-window, and all other control packets such as REVs and ACK-DATAs are scheduled to arrive at the target in its listen window and data packets are scheduled to arrive at the intended receiver in its reserved time slot. We discuss the scheduling algorithms next.

2.5 Scheduling at Sender

When a node has queued data packets and is in idle state, it then schedules to send a REV to the intended receiver so that the REV arrives in the receiver's listen window and, at the same time, avoids the R-windows of all its neighbors.

The sender first maps the whole listen window of the intended receiver and the R-windows of its neighbors into its own time line, then marks all the mapped R-windows which fall in the mapped listen window. After that, it divides the unmarked part (i.e., the available part) of the mapped listen window into slots by the duration of one control packet and randomly selects one slot as the time to transmit the REV. The REV specifies the required data duration and the offset of the to-be-reserved time slot to the beginning of its current period. When the sender calculates the duration needed to transmit the queued data packets, it has to count the time to skip the mapped R-windows of its neighbors.

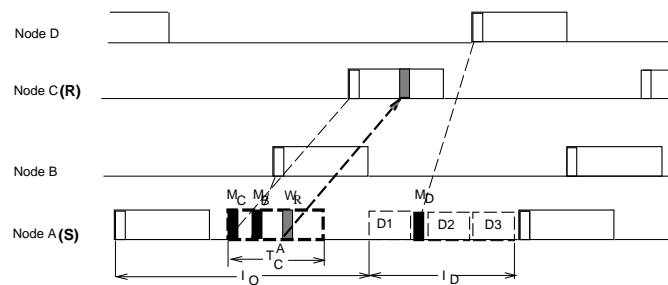


Figure 3. Scheduling at sender

Figure 3 illustrates the scheduling algorithm at the sender. In this figure, node A is the sender and node C is the intended receiver. Node A maps the listen window of node C , the R-windows of nodes C , B and D into its time line as T_C^A , M_C , M_B and M_D , respectively. When node A schedules to send a REV to node C , it randomly selects a slot from the unmarked part of T_C^A , denoted as W_R , to send the REV packet. The REV packet will arrive at node C during its listen window without interfering other nodes in their R-windows. In the REV packet, node A specifies the offset of the reserved time slot to the beginning of its current period, denoted as I_O , and the duration of the reserved time slot, denoted as I_D , which includes the transmission time of all the data packets ($D1$, $D2$ and $D3$ in the figure) and the time to skip the mapped R-window of node D , M_D .

The scheduling algorithm at the sender guarantees that no REV arrives in any node's R-window in its neighborhood. Furthermore, once the channel is granted, no data packet arrives in any node's R-window since the sender avoids all the R-windows of its neighbors when it transmits data packets.

2.6 Scheduling at Receiver

Once a reservation is selected, the receiver first schedules the transmission of ACK-REVs, and based on this schedule, arranges a time slot for the selected reservation. During the reserved time slot, the receiver powers on and waits for incoming data packets. After a pre-defined interval, if it does not receive any data packets as scheduled¹, it simply quits the receiving state and goes back to the periodic listen/sleep. After the receiver receives data packets in a burst, it schedules the transmission of ACK-DATA so that the ACK-DATA arrives in the sender's listen window.

2.6.1 Scheduling Algorithm for ACK-REVs

To avoid data transmission interference, the receiver should guarantee that before the sender receives its ACK-REV, all the receiver's neighbors have already been notified the reserved time slot. Therefore, when the sender receives the ACK-REV from the receiver, it is already granted the channel for the subsequent data transmission.

The receiver sends ACK-REV packets to its neighbors in their mapped R-windows to guarantee that these ACK-REVs arrive during their R-windows. However, the mapped R-window to send ACK-REV to the sender is the earliest mapped R-window which is greater than $S_i = M_i + 2 \times L_i$, i is any of the receiver's neighbors other than the sender, where M_i is the mapped R-window of node i at the receiver and L_i is the propagation latency from the receiver to node i . Here we introduce an additional one-way delay to handle the following case: after

¹It is possible that the sender receives another ACK-REV (from another pair of transmission) before the ACK-REV from this receiver, and thus can not speak. In this case, the sender has to back off and resend a REV later.

the receiver sends out an ACK-REV to one neighbor, it is possible that the neighbor transmits an ACK-REV (for another pair of transmission) to the receiver. In such case, the receiver checks if the reserved time slot specified in the incoming ACK-REV conflicts with its transmission of ACK-REVs. If yes, the receiver stops scheduling to send ACK-REVs. Otherwise, the receiver records the reserved time slot, continues to transmit its ACK-REVs and prepares for incoming data packets.

In some cases, the mapped R-windows at the receiver possibly overlap each other. In an even rare case, the receiver's own R-window overlaps with the mapped R-window of some neighbor. These special cases can be effectively handled in the second phase of R-MAC: period announcement. In the second phase, if a node finds out either one of these cases occurs, the node re-schedules its periodic sleep/listen. Since the duration of a R-window is very short compared with the duration of one period, for example, in our implementation of R-MAC, $T_R \leq 10ms$ and $T = 1s$, it only takes a few rounds for nodes to randomly select their schedules to avoid such cases.

2.6.2 Scheduling Algorithm for Reserved Time Slot

When the receiver determines the reserved time slot, it has to leave enough time for the ACK-REV to reach the sender and for the data packets to propagate from the sender to the receiver. Since the offset of the reserved time slot within a period of the sender is already specified in the REV packet, the receiver computes the offset of the reserved time slot to its own period and arranges the reserved time slot according to the transmission time of its ACK-REVs. The reserved time slot is the earliest time slot that is greater than $S_s = M_s + 2 \times L_s$, where M_s is the mapped R-window of the sender on the receiver and T_s is the propagation latency from the receiver to the sender. Therefore, when the sender receives the ACK-REV, it has enough time to deliver its data packets to the receiver in the reserved time slot.

In each ACK-REV packet, the receiver specifies the interval from the transmission time of this ACK-REV to the reserved time slot and the duration of the reserved time slot.

2.6.3 Scheduling Algorithm for ACK-DATA

When the receiver receives all the data packets, it schedules to acknowledge the data burst. R-MAC treats REV and ACK-DATA in the same way. That is, the receiver uses the same scheduling algorithm for REVs to schedule an ACK-DATA. It needs to make sure that the ACK-DATA arrives at the sender in its listen window. In the ACK-DATA packet, the receiver indicates whether a packet is received or corrupted by a bit vector.

2.6.4 Giving an Example

Now, we use an example to illustrate the scheduling at the receiver. Referring to Figure 4, again, node A is the

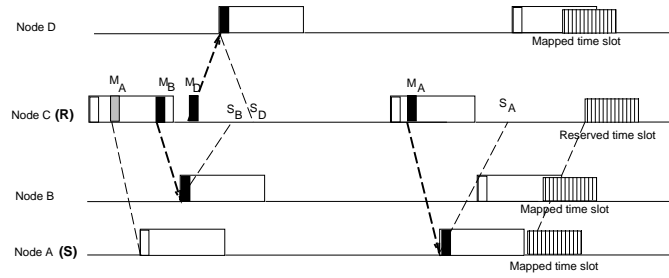


Figure 4. Scheduling at receiver

sender and node C is the receiver. M_A , M_B , and M_D are the mapped R-windows at node C for nodes A , B , and D respectively. Node C transmits ACK-REVs for node B and D first, then transmits ACK-REV for node A . S_D is the earliest time that M_A could be scheduled and S_A is the lower bound for the final reserved time slot.

2.7 Discussions

2.7.1 Fairness

Fairness is a desirable property for a MAC protocol. R-MAC supports fairness in two aspects. First, an intended receiver provides equal opportunity for all its neighbors to make reservations. Second, an intended receiver randomly selects one reservation from the reservations it collects.

However, in some network settings, for a given receiver, some nodes have advantage over others because of their locations and schedules. An example is shown in Figure 5, where node A is the intended receiver, nodes B and C have less propagation latency to node A than that of node D . When they compete to make reservations, nodes B and C always can schedule their REVs in the first period, while node D can only schedule its REV to arrive node A in the second period. Therefore, node B or C always wins the competition. In order to give equal opportunity to all neighbors, a node needs to collect reservations for several periods in a row. We call the required number of consecutive periods to collect reservations as *Collecting Interval*.

Clearly the collecting period is mainly determined by the period length. When the period length of nodes is too small, the collecting interval will most probably cover multiple periods. On the other hand, when the period length is big enough, say, greater than the time for a node to notify its neighbors of its reserved time slot (by scheduling ACK-REVs), which is bounded by $T_L + RTT$, then for any node who misses the first period for sending reservation can definitely catch the second period. Thus, we have the following lemma.

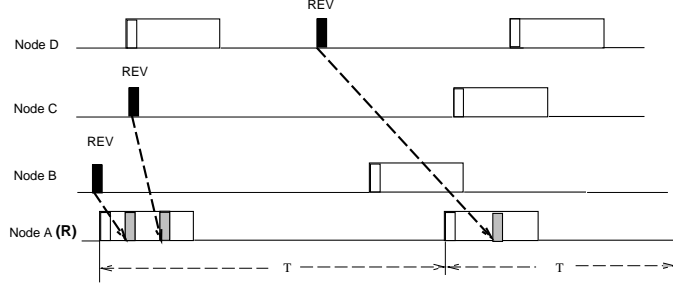


Figure 5. A fairness problem: nodes B and C win over node D

Lemma 1 *The collecting interval only needs two periods if the period length, $T_0 > T_L + RTT_{MAX}$, where RTT_{MAX} is the maximum RTT (Round Trip Time) between a sender and a receiver, and T_L is the listen window. $RTT_{MAX} = \frac{2 \times R}{V}$, where V is the sound speed in water and R is the maximum transmission range of the nodes in the network.*

Since T_L and RTT_{MAX} are usually small compared with T_0 (we will discuss how to set T_0 and T_L appropriately in Section 3), it is easy to satisfy the condition. Thus, in R-MAC, each node collects reservations for two consecutive periods. Once a node collects all the REVs, it randomly picks one of them and reserves a time slot for it. As we can see, R-MAC supports fairness at the cost of sacrificing channel utilization. In our targeted applications, however, channel utilization is not important compared with energy efficiency and fairness, as long as it satisfies the application requirements.

2.7.2 ARQ in R-MAC

Acknowledging each received packet may result in low channel utilization, high energy waste and high control packet overhead. Thus, in R-MAC, the receiver sends one acknowledgment packet (ACK-DATA) per data burst instead of per packet. In the ACK-DATA packet, the receiver indicates the lost packet by a bit vector. The length of the bit vector is the same as the maximum number of data packets in the burst. The value of a bit indicates whether the corresponding data packet is lost or received. We call this ARQ technique block-based acknowledgment.

It is possible that an ACK-DATA packet collides with other control packets. Consequently, the sender will retransmit the data packets. In order to avoid such waste, in R-MAC, the sender specifies the burst identification in the REV packet. Each node records the burst identifications and the most recent ACK-DATAs for all its neighbors. When a node receives a REV, it first checks the burst identification. If it is the same as the one recorded, it means that the sender does not receive the ACK-DATA packet, then it simply replies with an ACK-DATA. If the burst identification is different from the recorded one, it means that this is a new reservation. When the sender receives

the ACK-DATA, it updates its burst identification and clears the received data packets from its data queue. Since two values are enough for the burst identification per node, one bit in a REV is needed for the purpose of the burst identification.

2.7.3 Maintaining Synchronization

In R-MAC, nodes measure the propagation latencies to and the schedules of their neighbors in the first two phases. However, after a long time period, these measurements become more and more inaccurate because of the clock drift and varying sound speed in water. Therefore, the measurements of latencies and schedules need to be updated periodically.

There are two ways to keep these measurements accurate. One way is to piggyback a time stamp on each packet transmitted such as REV, DATA and ACK-REV. When the sender and the receiver exchange these packets, they update these measurements. Another way is to send control packets explicitly to update these measurements after a pre-defined time period.

Note that all the timestamps used in R-MAC are relative values, not absolute values. Thus the accuracy and stability of the synchronization depend on the frequencies of clocks. This greatly loosens the accuracy requirement on the node clocks. The typical relative drift of crystal-based clocks is in the order of 15-25 parts per million [3]. Such accuracy of the common crystal-based clocks is enough for R-MAC.

3 Analysis

In this section, we first analyze the properties of R-MAC. Then we explore several design parameters to get some guidelines for the real implementation of R-MAC.

3.1 Protocol Properties

From the schedule algorithms for REV, ACK-DATA, and DATA packets, we can see no REV, ACK-DATA or DATA packets arrive at a node's R-window. Only ACK-REVs can arrive in R-window of nodes. When a node receives ACK-REV in its R-window, it knows there exists a transmission in its neighborhood as well as the time duration of the transmission. In this way, the node could keep silent in that time period to avoid interrupting the transmission. When there are more than one ACK-REVs arriving at a node, i.e., ACK-REV packets collide, the node can sense the collision in its R-window and conclude that there are transmissions in the neighborhood, but it has no clue about time duration of the transmission. In this case, the node should back off for the worst case. Moreover, in R-MAC, all nodes keep listening in their R-windows to get the updated usage information of the

channel in the neighborhood. Therefore, R-MAC could well synchronize control and data packet transmissions so that even control packets collide, data packet collisions are still effectively prevented. Thus, we have the following theorem.

Theorem 1 *R-MAC guarantees data packets collision-free.*

In R-MAC, since all data packets are scheduled to skip R-windows of all nodes, if a node is exposed to a data sender, the R-window of this node will be skipped by the data sender. It means that this node can still send REV and receive ACK-REV without collision. Thus, the node is free to send data as long as the transmission does not interfere with the on-going data transmission. Therefore, we get the following theorem.

Theorem 2 *R-MAC solves the exposed terminal problem.*

3.2 Design Parameters

In R-MAC, there are several critical time intervals to be defined, such as the backoff time interval and the time interval that a sender needs to wait for an ACK-REV (referred to as *ACK-REV time interval* for short). These time parameters are determined by many factors such as the period length, the listen window, the maximum propagation latency and the maximum number of neighbors. In the following, we first discuss how to set period length and listen window appropriately, then we discuss how to choose the backoff time interval and ACK-REV time interval.

3.2.1 Setting Period Length T_0

Since the data transmission in R-MAC has to avoid all R-windows, we have minimum requirement on the period length. A very short period possibly causes the interval between any two mapped R-windows at a node too small, even less than the duration of one data packet, making R-MAC infeasible. We could give a lower bound for the period length T_0 as $D_P + N \times T_R$, where N is the maximum possible number of neighbors of a node, D_P is the duration of one data packet and T_R is the duration of the R-window. This lower bound gives the sufficient condition to guarantee that there exist at least one interval between any two consecutive mapped R-windows at a node, as allows the transmission of at least one data packet. To transmit C data packets in a burst, we have the following lemma.

Lemma 2 *In R-MAC, the period length T_0 has to be greater than $C \times D_P + N \times T_R$, where N is the maximum possible number of neighbors of a node, D_P is the duration of one data packet, C is the maximum number of data packets allowed in a burst and T_R is the duration of the R-window.*

3.2.2 Setting Listen Window

First, we define the minimum time period for any node used to collect REVs and ACK-DATAs as *contention window*, denoted by T_C . The probability that these control packets collide is determined by the durations of the control packets and contention window. A large contention window reduces the collision probability, but increases the energy consumption on idle state. We show the relation between the collision probability and the contention window as follows.

Let n be the number of REV or ACK-DATA packets a node receives in one period, D_C be the duration of a control packet (which is equal to T_R) and T_C be the listen window. We define p_i as the probability that i control packets are delivered successfully. We can easily get

$$\begin{aligned}
 p_1 &= 1 \\
 p_2 &= 1 - \frac{D_C}{T_C} \\
 p_3 &= p_2 \times \left(1 - \frac{2 \times D_C}{T_C}\right) \\
 &\dots \\
 p_n &= p_{n-1} \times \left(1 - \frac{(n-1) \times D_C}{T_C}\right)
 \end{aligned} \tag{1}$$

Then we get the following equation

$$p_n = \left(1 - \frac{(n-1) \times D_C}{T_C}\right) \left(1 - \frac{(n-2) \times D_C}{T_C}\right) \dots \left(1 - \frac{D_C}{T_C}\right) \tag{2}$$

Equation 2 gives the probability that n control packets can be delivered successfully in the contention window T_C . Equation 2 assumes that the contention window at all nodes are slotted in the same way. This is skewed in the real implementation since the listen windows (therefore, contention windows) are slotted differently at different nodes because of the different R-window mapping at these nodes. But it is still a close estimation of the relation between the REV/ACK-DATA success probability and contention window.

Given T_0 , D_C , n and the requirement for the probability that n control packets are successfully delivered, we can set the listen window accordingly. For example, if $n = 3$ and $D_C = 0.005s$, if we want the success rate of control packets is greater than 90%, then $T_C = 0.2s$ can guarantee such probability requirement.

Further, we use N_r to denote the maximum number of mapped R-windows allowed in a node's listen window. Then the listen window T_L should as follows: $T_L > (1 + N_r) \times T_R + T_C$. We can also have N_r is statistically equal to $N \times \frac{T_L}{T_0}$, where N is the maximum number of neighbors and T_0 is the period time. Thus, we have $T_L > (1 + N \times \frac{T_L}{T_0}) \times T_R + T_C$, from which we can set T_L appropriately.

3.2.3 Backoff Time Interval and ACK-REV Time Interval

When a node senses a collision in its R-window, this node knows that the channel is reserved but does not know how long the reservation takes. Therefore, the node has to prepare for the worst case, it keeps silence for the time interval (i.e., the backoff time interval) that is long enough for any data transmission. If the length period is greater than $\max(T_L + RTT_{MAX}, N \times T_R + C \times D_P)$, i.e., satisfying the minimum length and the basic fairness requirement, the maximum number of periods a transmission takes at most 5 periods: up to 2 periods for REV transmissions, up to 2 periods for ACK-REV transmissions, and 1 period for data transmission. Then we have the following lemma.

Lemma 3 *When the period length, T_0 is greater than $\max(T_L + RTT_{MAX}, N \times T_R + C \times D_P)$, then a data transmission takes at most 5 periods, where RTT_{MAX} is the maximum RTT in R-MAC, T_L is the listen window, N is the maximum number of neighbors, D_P is the duration of one data packet, C is the maximum number of packets allowed in a burst and T_R is the duration of the R-window.*

Therefore, the upper bound of backoff time interval is 5 periods. Similarly, we can get a upper bound of 4 periods for ACK-REV time interval.

4 Simulation Study

In this section, we evaluate the performance of R-MAC using simulations. By comparing with T_u -MAC (a revised version of T-MAC for underwater sensor networks), we demonstrate the energy efficiency of R-MAC. We also conduct experiments to explore the fairness and channel utilization of R-MAC.

4.1 Simulation Settings

We implement R-MAC in Aqua-Sim, an NS-2 based simulator for underwater sensor networks, developed at the Underwater Sensor Network (UWSN) Lab at the University of Connecticut (<http://uwsn.engr.uconn.edu/>).

Unless specified otherwise, we use the following parameters in the simulations. We set the size of control packets to 5 Bytes and the data packet size to 60 Bytes. The period length is 1 second and the listen window is 100 ms. The bit rate is 10 Kbps. The data generation follows a Poisson process with an average rate as λ , called *traffic rate*. The maximum transmission range is 90 meters. The interference range is the same as the transmission range. The maximum number of data packets allowed in a burst is 3. In our simulations, we set the energy consumption parameters based on a commercial underwater acoustic modem, UMW1000, from LinkQuest [9]:

the power consumption on transmission mode is 2 Watts; the power consumption on receive mode is 0.75 Watts; and the power consumption on sleep mode is 8 mW.

4.1.1 Performance Metrics

We define four metrics to quantify the performance of R-MAC: working time, overhead, goodput and end-to-end delay.

Working time is the sum of time that the nodes in the network spend on transmitting, receiving or idle state. We have transmitting time, receiving time and idle time depending on the state we are interested in.

Overhead is the ratio of working time to the number of packets successfully received by the receiver. Corresponding to the types of working time, we have transmitting overhead, receiving overhead and idle overhead. Overhead is the metric of energy efficiency. Since the energy consumption is determined by the implementation of modems, it is more generic to use working time and overhead as metrics for energy efficiency.

Goodput is the number of packets successfully received by the receiver. We use goodput as the measurement for the channel utilization.

End-to-end delay is the time interval from the time that a packet is delivered to the MAC to the time that this packet arrives at the destination node.

4.2 Implementation of T_u -MAC

As we mentioned earlier, there are no energy efficient MAC solutions for underwater sensor networks with unevenly distributed traffic in the literature. We choose to implement T-MAC [15] as a reference because both T-MAC and R-MAC can adapt to unevenly distributed traffic with high energy efficiency, though T-MAC is designed for radio-based sensor networks. We apply the idea of T-MAC in underwater sensor networks to show that it is not efficient to simply adapt radio-based network protocols in underwater network environment.

To make T-MAC feasible for underwater sensor networks, we implement a revised version of T-MAC, referred to as T_u -MAC. We make three major revisions: First, we modify the active time, TA , to incorporate propagation delay, which is non-negligible in underwater sensor networks; Second, we modify the RTS/CTS method adopted in T-MAC. As shown in [17], the original RTS/CTS method is not suitable for underwater sensor networks due to long propagation delays. In order to make it practical in underwater sensor networks, we make the following change: when the sender receives a CTS from the intended receiver, it has to wait until the CTS propagates through the whole transmission range of the receiver; Third, since carrier sensing does not make much sense in underwater sensor networks, T_u -MAC does not adopt this technique. Instead, it has the following design: if a node senses

the channel busy when it starts to transmit a packet, it will back off for a random time interval between one data packet duration and two data packets durations.

4.3 Fairness of R-MAC

In this set of simulations, we investigate the performance of R-MAC to share medium. We check the goodputs of all the contender under different traffic rates.

We use the topology shown in Fig. 6. In the simulations, nodes 1, 2, 3 and 4 send data to node 0. The distances between the senders and the receiver are different. We vary the data sending rate from 0.02 to 0.24 packets per second. We measure the goodput and energy consumption of each sender. The results are shown in Fig. 7 and Fig. 8 with standard deviations.

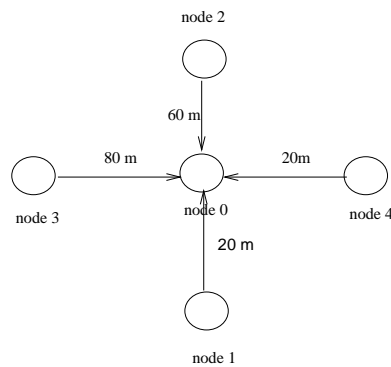


Figure 6. The topology used to evaluate the fairness of R-MAC

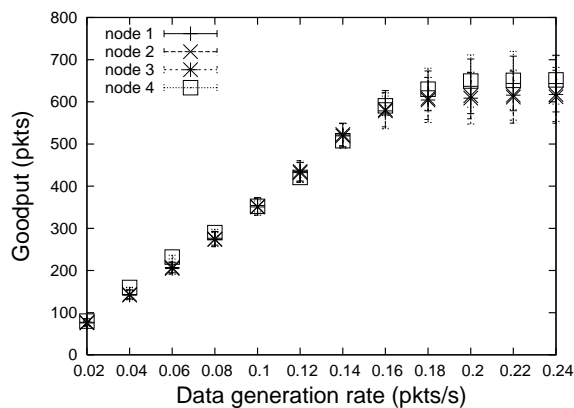


Figure 7. Goodput of R-MAC

As shown in Fig. 7, all the senders have a very close share of the channel regardless of their positions. The goodputs of the senders increase along with the traffic rate. However, when the traffic rate reaches some threshold

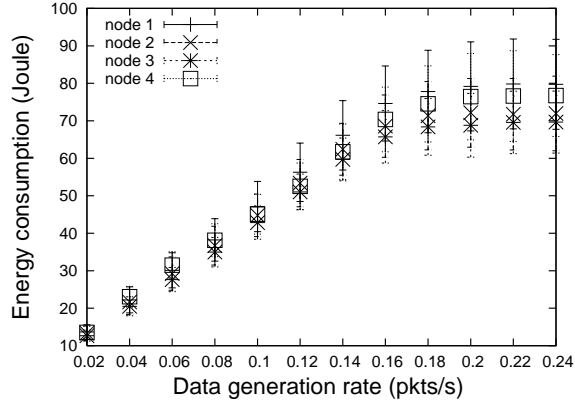


Figure 8. Energy consumption of nodes in R-MAC

(0.2 packet/s in these simulations), the senders saturate the channel, and the goodputs become flat and do not change with the traffic rate since then. The curves in Fig. 7 also shows the effect of collision avoidance provided by R-MAC, i.e., when the traffic rate is low, all the data packets generated get through. However, when the traffic rate exceeds the capacity of the channel, only some data packets are sent, the traffic rate has no effect on the goodput anymore.

The energy consumption of each sender with the standard deviation is shown in Fig. 8. As shown in this figure, these senders also have the same energy consumption under different traffic rates.

From these figures, we can see that all the senders in R-MAC have close goodputs at close energy consumption when they contend for the channel. R-MAC supports fairness in various traffic rates.

4.4 Energy Efficiency

We evaluate the energy efficiency of R-MAC and T_u -MAC using a star topology as shown in Figure 9. In this network, node 0 is the only receiver and all other nodes are senders. All the senders can hear each other. We measure the energy consumption of R-MAC and T_u -MAC at different traffic rates. To make the comparison fairly, we choose the same parameters for R-MAC and T_u -MAC wherever this rule applies. It should be also noted that the ARQ technique adopted in the original T-MAC is different from that used in R-MAC. We thus also implement ARQ per burst, i.e., burst-based acknowledgement, for T_u -MAC.

To analyze how R-MAC achieves energy efficiency, in the following we first measure the average time spent on transmitting, receiving, and idle state per successfully delivered packet, for each we call transmitting overhead, receiving overhead and idle overhead respectively.

The results for transmitting overhead are shown in Figure 10. From this figure, we can see that the transmitting

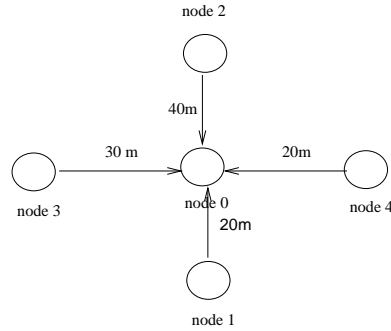


Figure 9. Star topology for energy efficiency

overhead of R-MAC is higher than that of T_u -MAC when the traffic rate is low, but it becomes less as the traffic rate continues to increase. This is because that the receiver in R-MAC has to transmit an ACK-REV to each of its neighbors, while the receiver in T_u -MAC just needs to send one CTS for all of its neighbors. Thus when the traffic rate is low, the number of packets in a burst is small, which means that the probability that data packets in T_u -MAC collide is very low. Thus the transmitting overhead of R-MAC is higher than that of T_u -MAC. As the traffic rate keeps increasing, the number of data packets in a burst becomes larger, which leads to the reduction of transmitting overhead in R-MAC. On the other hand, when the data burst size is lifted, the data packet collision in T_u -MAC becomes more and more significant, which causes the increase of transmitting overheads in T_u -MAC. When the traffic rate is relatively high (greater than 0.14 pkts/s as shown in the figure, the data packet collision is the dominating source for the transmitting overheads in T_u -MAC. Therefore, in this case, the transmitting overhead of R-MAC is lower than that of T_u -MAC.

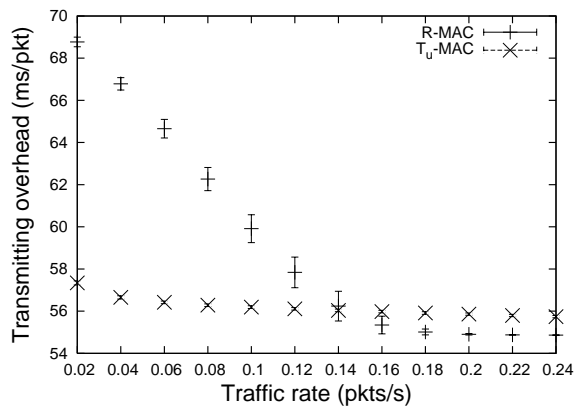


Figure 10. Transmitting overhead

The results for receiving overhead are shown in Figure 11. As we can see, the receiving overhead for R-MAC is only half that of T_u -MAC. Such significant difference on the receiving overhead is attributed to the random

schedule in R-MAC. In T_u -MAC, all nodes listen at the same time; therefore, when a node sends packets, all its neighbors overhear these packets, resulting in higher receiving overhead. On the other hand, in R-MAC, nodes have different schedules. When a node sends packets, it is most likely that other nodes are in their sleep state.

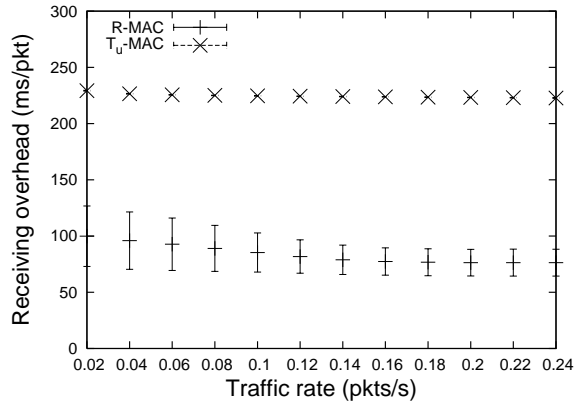


Figure 11. Receiving overhead

Figure 12 shows the results of idle overhead. From this figure, we can observe that idle overhead decreases with the growth of traffic rate and the idle overhead of R-MAC is only half of that of T_u -MAC. This is mainly caused by two factors. First, the method adopted in T_u -MAC to reduce end-to-end delay results in longer idle time on each node. In T_u -MAC, nodes have to stay idle for possible future traffic when they find the channel is captured by other nodes. Second, the long active time, TA, of each node also contributes to the high idle overhead of T_u -MAC.

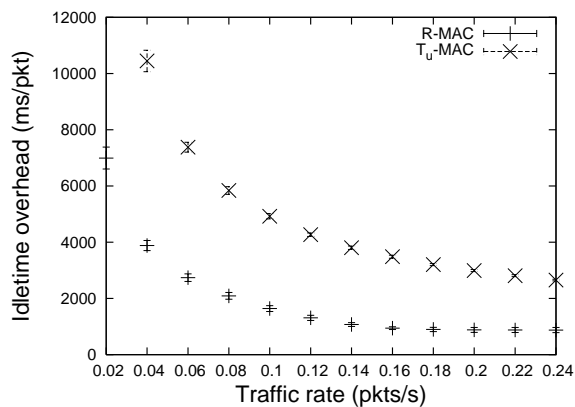


Figure 12. Idle overhead

We convert the overhead shown in Figure 10, Figure 11 and Figure 12 into energy consumption per data packet based on the parameters of UMW1000 modems, and the results are plotted in Figure 13. From this figure, we can observe that under low traffic rate such as 0.02 pkts/s, even R-MAC has higher transmission overhead (per data

packet), it is still more energy efficient than T_u -MAC due to its lower receiving overhead and less idle overhead per data packet. As the traffic rate increases, the difference between R-MAC and T_u -MAC is enlarged. This is mainly due to the transmitting inefficiency of T_u -MAC when the traffic rate is high.

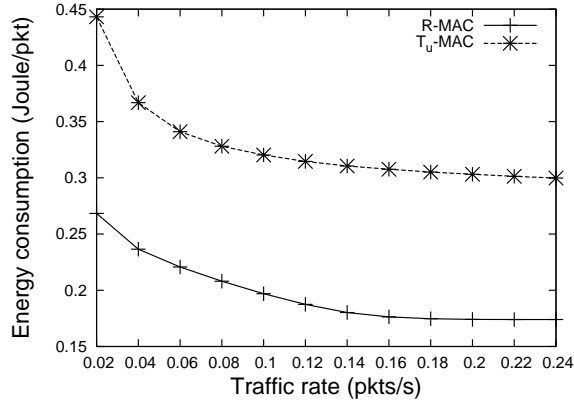


Figure 13. Energy consumption

From this set of simulations, we can conclude that R-MAC is much more energy efficient than T_u -MAC under various traffic rates. This tells us that we can not directly apply T-MAC, a well designed MAC solution for radio-based sensor networks, in underwater network scenarios. The conclusion also indicates that R-MAC can achieve high energy efficiency under unevenly distributed traffic.

4.5 End-to-End Delay

In this set of simulations, we evaluate the end-to-end delays of R-MAC and T_u -MAC in a multi-hop network. The topology is shown in Figure 14, where node 0 is the receiver and node n is the only sender. The distance between any two adjacent nodes is 80 meters. The sender sends data at the average rate of 0.1 packet per second (a light traffic rate). We vary the number of hops by adding different number of nodes between the sender and the receiver. For each number of hops, we compute the average end-to-end delay of all the packets received by the receiver.

The end-to-end delay results are plotted in Figure 15. From this figure, we can see that the end-to-end delay of both R-MAC and T_u -MAC increases linearly as the number of hops increases. These results are consistent with our analysis : at light traffic, R-MAC at most takes $5T_0$ to transmit one packet per hop, and T_u -MAC takes about T_0 . This is mainly due to two factors: 1) T_u -MAC explicitly introduces a technique to reduce end-to-end delay (as described earlier) with the cost of sacrificing energy efficiency; 2) R-MAC introduces additional delay (one period) to guarantee fairness. In other words, R-MAC trades end-to-end delay for energy efficiency and fairness.

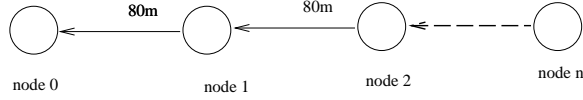


Figure 14. Line topology for end-to-end delay

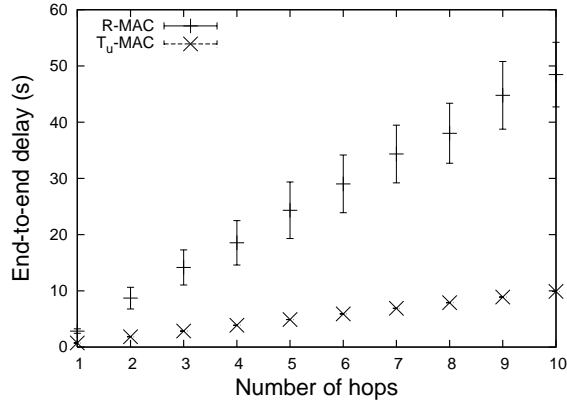


Figure 15. End-to-end delay when traffic rate is 0.1pkt/s

We also run simulations for high traffic rate. In fact, due to the good scheduling of R-MAC, the increase of end-to-end delay in R-MAC is much smaller than the increase in T_u -MAC. For example, for a traffic rate of 1 packet per second (referring to Figure 16), the end-to-end delay of T_u -MAC for 10-hops is around 100 seconds, while the result of R-MAC is around 300 seconds. As we increase the traffic rate to very high, the end-to-end delays of R-MAC and T_u -MAC are very close. This is because the packet queueing delay is dominating when the transmission competition is very intensive. Moreover, the delivery ratio of T_u -MAC drops much more significantly at high traffic than that of R-MAC. Thus, if we combine the delay results with the energy results, we can conclude that R-MAC has more advantages in the networks with high data traffic.

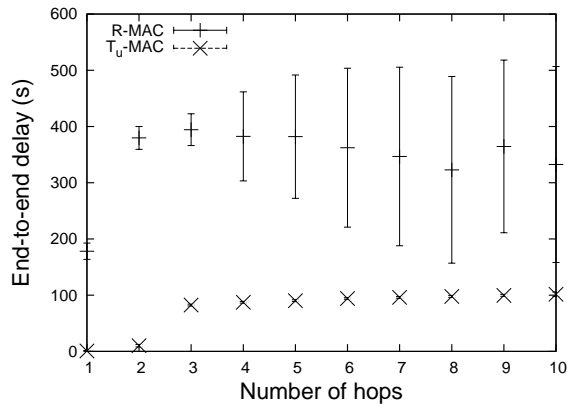


Figure 16. End-to-end delay when traffic rate is 1pkt/s

5 Related Work

Many widely used MAC protocols, such as IEEE 802.11, for radio networks are based on the RTS/CTS approach. The RTS/CTS approach was first proposed in MACA[8]. Then a variant protocol, MACAW, was proposed in [4]. This protocol adopts backoff and ARQ techniques in addition to the RTS/CTS control message exchange. Later, carrier sensing was combined with RTS/CTS in a new protocol, called FAMA, in [6].

A major problem with these RTS/CTS based protocols is energy efficiency: too much energy is wasted in idle state since all nodes keep powered on all the time.

With the emerging of sensor networks, energy-efficient MAC becomes a hot topic. PAMAS proposed in [14] makes an improvement to save energy by putting nodes into sleep state when the nodes are prohibited from sending any packet. PAMAS improves MACA without sacrificing the throughput and end-to-end delay. However, in order to turn off/on the nodes intelligently, PAMAS uses a separate signaling channel as a control channel, as is not desirable in sensor networks.

S-MAC proposed in [18] uses the in-channel signal to control the node to listen and sleep periodically. S-MAC reduces energy consumption on listening significantly. However, the fixed duty cycle in S-MAC is undesirable since the traffic in sensor networks varies with locations and time. T-MAC proposed in [15] improves S-MAC by adopting an adaptive duty cycle scheme to reduce the energy consumption while maintaining a reasonable throughput.

Our R-MAC design has benefited from the previous techniques in various aspects. However, as shown in Section 4, we cannot directly apply radio-based ideas into underwater network scenarios. Instead, new MAC solutions are required for underwater sensor networks. In the literature, there are only few MAC protocols proposed for underwater networks. We discuss two recent proposals as follows.

A random access based MAC protocol for underwater sensor networks is proposed in [13]. Due to the fundamental limitation of random access, this protocol only works well for the networks with very low and evenly distributed traffic. Moreover, this protocol has high overhearing and idle waste since for a given node all its neighbors have to wake up for the possible traffic, even when this node sends nothing.

Recently, a modified FAMA, called slotted FAMA, is proposed in [10] for underwater acoustic networks. In slotted FAMA, time is divided into slots, and all packets including control packets and data packets are sent at the beginning of a slot. In this way, the lengths of RTS and CTS are not determined by the propagation delay as is in the original FAMA, thus making the protocol feasible for underwater acoustic networks. It should be noted that energy efficiency is not the design goal of slotted FAMA.

6 Conclusions and Future Work

In this paper, we propose an energy efficient MAC protocol, R-MAC, for underwater sensor networks. R-MAC carefully schedules the transmissions of control and data packets to avoid data packet collisions. The scheduling algorithms not only avoid data packet collisions completely, but also solve the exposed terminal problem. In R-MAC, each node adopts periodic listen/sleep to reduce the energy waste in idle state and overhearing. Moreover, R-MAC loosens the synchronization requirement by allowing each node randomly selects its own schedule. Additionally, R-MAC supports fairness. Finally, the burst-based acknowledgment technique reduces the control packet overhead further and improves the channel utilization.

We plan to integrate some coding schemes in the burst-based acknowledgment to make R-MAC more robust and energy efficient under error-prone communication channels. Considering the harsh underwater environments, we believe that the combination of coding and ARQ is promising for reliable underwater acoustic communications.

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