# Token-Based Adaptive MAC for a Two-Hop Internet-of-Things Enabled MANET 

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#### Abstract

In this paper, a distributed token-based adaptive medium access control (TA-MAC) scheme is proposed for a twohop Internet of Things (IoT)-enabled mobile ad hoc network. In the TA-MAC, nodes are partitioned into different one-hop node groups, and a time division multiple access (TDMA)-based superframe structure is proposed to allocate different TDMA time durations to different node groups to overcome the hidden terminal problem. A probabilistic token passing scheme is devised to distributedly allocate time slots to nodes in each group for packet transmissions, forming different token rings. The distributed time slot allocation is adaptive to variations of the number of nodes in each token ring due to node movement. To optimize the medium access control (MAC) design, performance analytical models are presented in closed-form functions of both MAC parameters and network traffic load. Then, an average end-to-end delay minimization framework is established to derive the optimal MAC parameters under a certain network load condition. Analytical and simulation results demonstrate that, by adapting the MAC parameters to the varying network condition, the TA-MAC achieves consistently minimal average end-to-end delay, bounded delay for local transmissions, and high aggregate throughput. Further, the performance comparison with other MAC schemes shows the scalability of the proposed MAC in an IoT-based two-hop environment with an increasing number of nodes.


Index Terms-Adaptive medium access control (MAC), aggregate throughput, end-to-end delay, Internet of Things (IoT), multihop mobile ad hoc network (MANET), scalability, time division multiple access (TDMA), token passing.

## I. Introduction

THE INTERNET of Things (IoT) has great potentials to be one of the most promising network infrastructures toward the next generation wireless network evolution. The IoT framework will interconnect a growing number of heterogeneous objects, i.e., smartphones, sensors and actuators, autonomous devices, via suitable wireless technologies for ubiquitous Internet access and pervasive information sharing [1]-[3]. Within this framework, various IoT-oriented intelligent applications can be realized, e.g., disaster monitoring and response, intelligent control for smart homing, and industrial automation. To support an increasing node number and user demands, an IoT-enabled mobile ad hoc network (MANET) emerges as an

[^0]important means to provide seamless Internet access for end users. ${ }^{1}$ A MANET consists of a group of self-organized nodes, interconnected for communication in a peer-to-peer manner, without any centralized control. Due to low cost and simplified implementation, MANETs are widely deployed for applications such as smart home networking [1], emergency communications, and prompt response in postdisaster areas [4]-[6].
For an IoT-enabled MANET, to maintain consistently satisfactory performance in presence of network traffic load variations due to node mobility, an efficient medium access control (MAC) protocol is imperative to coordinate packet transmissions of each node in a distributed way and to adapt to the network traffic load variations [7]. However, the distinctive characteristics of IoT pose new technical challenges on MAC for MANETs.

1) The IoT infrastructure should accommodate an increasing number of users. For example, in disaster-affected areas without conventional communication infrastructures, an increasing number of smart devices from victims can be connected via ad hoc networking to support an abrupt rise of data traffic and communication demands after the catastrophe. Therefore, the MAC protocol should be scalable to the number of nodes to achieve high network throughput and low transmission delay, especially under high network load conditions [8].
2) The increased number of nodes can enlarge the network coverage area, making the communication distance between a pair of end users beyond the one-hop transmission (communication) range.
3) For a multihop network, some nodes staying in the transmission ranges of both source and destination nodes (that are far apart) may relay traffic for the end nodes. Thus, the compound traffic arrival rate (superposition of the external traffic arrival rate and the relay traffic arrival rate) at each relay node can become high, resulting in a large overall delay for relay transmissions and thus for end-to-end transmissions. Therefore, how to maintain a consistently minimized end-to-end packet delay in a multihop environment with an increased number of nodes is critical for MAC.
In literature, contention-based carrier sense multiple access with collision avoidance (CSMA/CA) using request-to-send/clear-to-send handshaking schemes, e.g., IEEE 802.11 MAC [8], [9], has been demonstrated not scalable in a high network load condition in a multihop environment,

[^1]due to increased transmission collisions caused by the hidden terminal problem [10] and/or the receiver blocking problem [11], which become worse with an increasing number of nodes. The above problems can be solved by a dual-channel busy-tone-based MAC solution [12] at the price of increased protocol complexity and additional circuitry [11]. Time division multiple access (TDMA) protocols [13], [14] perform better for multihop transmissions, achieving high channel utilization by eliminating unintentional packet collisions due to the hidden terminal problem. In [13], a joint TDMA-based MAC and routing protocol is proposed for packet transmissions in a multihop vehicular ad hoc network (VANET), in which every vehicle can acquire a time slot not occupied by any of its two-hop neighbors upon listening to the neighboring information exchange within each frame. Dynamic TDMA time slot assignment (DTSA) is presented in [15] to support a varying number of users in a multihop MANET, where the frame length is doubled each time when no time slots are available for newly arriving nodes in current frame. Recently, hybrid MAC protocols, combining CSMA/CA with TDMA, are revisited for a multihop environment to achieve a performance tradeoff between the two MAC approaches, which can be effective in a low load condition, for example, the unused transmission time slot contention [16] and the CSMA/CA-based time slot scheduling [17]. However, the network scalability is still throttled due to contention collision accumulation in a high load condition. Token-based MAC protocols, as a subset of contention-free protocols, have also gained many research interests for MANETs, due to its quality-of-service (QoS) provisioning capability [18]-[20] and the flexibility in supporting network topology changes [21]. A multichannel token ring-based MAC protocol is proposed in [22] for supporting both safety and nonsafety packet transmissions in a multihop VANET, where inter-ring communications are based on the CSMA/CA and token passing is employed for intraring data communications. In [23], a dualchannel token-based MAC protocol is proposed for multihop MANETs, which have a control channel for token passing and channel reservation, and a data channel for data transmissions. The performance analysis is carried out for a single-hop scenario.

The end-to-end packet delay is an important performance metric to reflect the effectiveness of a MAC protocol in a multihop environment. However, most of the existing TDMA and token-based protocols [22], [23] allocate time slots and schedule the token passing without considering the end-toend delay satisfaction due to the intractability of analytical modeling for the end-to-end delay and its optimization in a multihop network. Thus, the end-to-end packet delay can increase to an unacceptable level with an increasing node number, if transmission opportunities are not adaptively allocated. Therefore, adapting the allocation of TDMA time slots or the scheduling of token rotation cycles to the network traffic load variations is of paramount importance to ensure the protocol scalability, with a low end-to-end delay and a high aggregate network throughput. In this paper, we consider a two-hop network as the first step toward a more general multihop environment, and propose a token-based adaptive

MAC (TA-MAC) scheme. In the TA-MAC, both the number of token rotation cycles and the superframe duration are optimized and adapted to the instantaneous network traffic load, to achieve a consistently minimal average end-to-end packet delay. Our contributions are threefold.

1) First, to eliminate the hidden terminal problem, a distributed TDMA-based superframe structure is considered for the TA-MAC, in which different one-hop node groups are allocated different TDMA durations. Inspired by Liang and Zhuang [19] and Wang and Zhuang [20], each node group forms a token ring by adopting a probabilistic token passing scheme to distributedly allocate time slots to the group members for packet transmissions. Each token ring maintains and updates its node members in a distributed way, and the transmission time slot allocation is adaptive to the instantaneous number of nodes in the network.
2) Second, to determine the MAC parameters for performance optimization, we evaluate the average delay for end-to-end packet transmissions, the average delay for local packet transmissions, and the aggregate network throughput for the TA-MAC in closed-form functions of the MAC protocol parameters and the network traffic load.
3) Third, with a predefined superframe length, an optimization framework is established for minimizing the average end-to-end delay under the constraints of guaranteeing the bounded average delay for local transmissions and maintaining stable transmission queues of each node. The original nonconvex minimization problem is then decoupled into a convex subproblem and a biconvex subproblem, which can be solved sequentially to obtain the minimized number of token rotation cycles for each token ring. Then, a distributed computation algorithm is proposed to determine the optimal superframe length and the associated optimal numbers of token rotation cycles for each token ring, with which the minimal average end-to-end delay can be achieved.
This paper is organized as follows. The system model is described in Section II. In Section III, the TA-MAC scheme is presented to support packet transmissions from a varying number of mobile nodes in the two-hop MANET. Section IV provides the performance analysis of the proposed MAC scheme, where the average end-to-end delay is derived as a closed-form function of a set of MAC parameters. An end-toend delay optimization framework is established in Section V to obtain the MAC parameters, with which the minimal average end-to-end delay can be achieved. Extensive analytical and simulation results are presented in Section VI to show the scalability of the proposed MAC scheme in supporting two-hop packet transmissions in a high traffic load condition. Finally, we draw the conclusions in Section VII. Main parameters and variables are listed in Table I.

## II. System Model

For a multihop MANET, the communication distance between a pair of source-destination (S-D) nodes can be

TABLE I
Main Parameters and Variables

| Parameter \& Variable | Definition |
| :---: | :--- |
| $D_{a c}\left(D_{b c}\right)$ | Average delay for packet transmissions from area A (B) to C |
| $D_{c a}\left(D_{c b}\right)$ | Average delay for relay packet transmissions from area C to A (B) |
| $D_{a}\left(D_{b}\right)$ | Average delay for local packet transmissions within area A (B) |
| $D_{a b}\left(D_{b a}\right)$ | Average end-to-end delay for the transmission direction from area A (B) to B (A) |
| $D^{*}$ | Minimized average end-to-end delay given a certain superframe length |
| $D^{o p t}$ | Minimal average end-to-end delay |
| $D_{t h}$ | A delay bound for local packet transmissions within area A and area B |
| $k_{j}$ | Number of token rotation cycles scheduled for token ring $R_{j}(j=a c, b c, a, b)$ |
| $k_{j}^{*}$ | Optimal number of scheduled token rotation cycles given a certain superframe length |
| $k_{j}^{o p t}$ | Optimal number of scheduled token rotation cycles |
| $L_{j}$ | Number of nodes in token ring $R_{j}(j=a c, b c, a, b)$ |
| $M^{\prime}$ | Total number of time slots in each superframe |
| $M^{o p t}$ | Optimal total number of time slots for each superframe |
| $N$ | Total number of nodes in the two-hop network |
| $N_{a}, N_{b}, N_{c}$ | Number of nodes in network areas A, B, C |
| $R_{a c}, R_{b c}, R_{a}\left(R_{b}\right)$ | Four token rings among different node groups |
| $T_{a c}, T_{b c}, T_{a b}$ | Three TDMA time durations in each superframe |
| $T_{f}$ | Superframe length |
| $T_{f}^{o p t}$ | Optimal superframe length |
| $T_{s}$ | Transmission time slot duration |
| $T_{1} / T_{2}$ | Idle duration before data (token) packet $/$ REQUEST packet transmissions |
| $W_{s, j}$ | Packet service time for a node in token ring $R_{j}(j=a c, b c, a, b)$ |
| $\lambda$ | External traffic arrival rate at each node |
| $\lambda_{a c}\left(\lambda_{b c}\right)$ | External traffic arrival rate heading to area C from nodes in area A (B) |
| $\lambda_{c a}\left(\lambda_{c b}\right)$ | Compound traffic arrival rate at a relay node for a destination in area A (B) |
| $\lambda_{a}\left(\lambda_{b}\right)$ | External traffic arrival rate at nodes in area A (B) for local transmissions |
| $\mu_{j}$ | Packet service rate for nodes in $R_{j}(j=a c, b c, a, b)$ |
| $\mathcal{L}(j)$ | Probabilistic token passing list for token ring $R_{j}(j=a c, b c, a, b)$ |
| $\mathcal{N}(x)$ | One-hop neighbor node IDs of node $x$ |
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|  |  |
|  |  |

larger than the one-hop transmission range. Therefore, some intermediate relay ( $R$ ) nodes, residing within both transmission ranges of the S-D nodes that are not reachable to each other directly, not only transmit data packets from their own application layers, but may also relay packets between the S-D node pair. Fig. 1(a) illustrates a general MANET, where each dashed circle represents a fully connected network (a onehop cluster), i.e., nodes in the network are within the one-hop communication range of each others. Some nodes, denoted by black dots, staying in the overlapping areas of different onehop clusters can act as relays to forward packets for other S-D node pairs (denoted by white dots) beyond the direct communication range. Therefore, the basic communication unit in a multihop MANET is a two-hop network, and some nodes can be $R$ nodes in addition to S-D nodes. With mobility, nodes can leave one two-hop network and become members of another one. In this paper, we consider a basic two-hop network model as the first step toward a general multihop environment, shown in Fig. 1(b). There are three logical areas $(A, B$, and $C)$. Nodes enter or depart from the network coverage region, or move around in the three areas. For a given packet transmission direction, such as from left to right, a node can be an $S(D)$ node or an $R$ node depending on whether its location is in area $A(B)$ or $C$.

Let $N$ denote the total number of nodes in the network which can slowly vary with time due to node mobility, and $N_{a}, N_{b}$, and $N_{c}$ denote the numbers of nodes in areas $A, B$, and $C$, respectively. There is a single type of data traffic in the network. The compound traffic arrivals for each $R$ node


Fig. 1. (a) General multihop MANET. (b) Simplified two-hop network.
are different from those of an $S(D)$ node, which consists of not only the traffic arrivals generated from its own application layer but also the relay traffic coming from nodes in both areas $A$ and $B$ [13]. Packet arrivals at each node are split into two traffic streams according to different transmission directions: an arriving packet at each $S$ node in area $A(B)$ is transmitted either to a local $D$ node in the same area or to an end $D$ node two-hop away in area $B(A)$. Similarly, each $R$ node transmits its self-generated packets to a $D$ node in either areas $A$ or $B$, and also relays packets from area $A(B)$ to a $D$ node in area $B(A)$.

There is a single radio channel in the network, without transmission errors. Nodes access the channel in a distributed manner. We assume that each node is equipped with a global positioning system (GPS) receiver, and the time synchronization among nodes in the network can be achieved by using the 1PPS signal provided by any GPS receiver [24]. Transmission failures can happen due to packet collisions, i.e., more than


Fig. 2. Superframe structure.


Fig. 3. Packet transmissions in four one-hop subnetworks during (a) $T_{a c}$, (b) $T_{b c}$, and (c) $T_{a b}$.
one transmission attempts are initiated simultaneously by different nodes. Each node has an exclusive node identifier (ID) that can be selected at random and included in each transmitted packet [24]. For a tagged node $x$, we denote the set of node IDs of all one-hop neighbors of node $x$ (including $x$ itself) as $\mathcal{N}(x)$. In the network, time is partitioned into a sequence of superframes, and the length of each superframe, denoted by $T_{f}$, is determined based on the numbers of nodes in the network areas. As shown in Fig. 2, we partition each superframe into durations, $T_{a c}, T_{b c}$, and $T_{a b}$, which consist of, $M_{a c}, M_{b c}$, and $M_{a b}$, numbers of time slots of equal duration $T_{S}$, respectively. Therefore, the duration of each superframe, $T_{f}$, is equal to $M \cdot T_{S}$, where $M$ is the total number of time slots within a superframe. Each time slot can accommodate one data packet transmission, and nodes transmit packets at the start of each time slot. To resolve the hidden terminal problem [14], [25] for the two-hop network, durations $T_{a c}$ and $T_{b c}$ are reserved for communications between nodes in areas $A$ and $C$ and between nodes in areas $B$ and $C$, respectively, where the transmitting and receiving nodes of a communication pair are in different areas, as shown in Fig. 3(a) and (b). The last duration, $T_{a b}$, is reserved for simultaneous communications among nodes in area $A$ and among nodes in area $B$, where both transmitting and receiving nodes are in the same area, as shown in Fig. 3(c). With this spatial reservation of transmission time slots for the four node groups forming four one-hop subnetworks, packet collisions caused by hidden terminals can be completely eliminated.

## III. TA-MAC Scheme

## A. Probabilistic Token Passing Within Each Node Group

In the TA-MAC, there are four tokens circulated separately among nodes in each group for packet transmissions, forming four token rings, $R_{a c}, R_{b c}$, and $R_{a}\left(R_{b}\right)$. For each token ring, when a node holds a token, it is assigned a time slot with duration $T_{s}$ for transmission of either a data packet ${ }^{2}$ or a token packet [22], [23], and the current token holder decides which

[^2]node is the next token holder. We define one token rotation cycle as the time duration a token has visited all node members once in token ring $R_{j}(j=a c, b c, a, b)$, which equals $L_{j} \cdot T_{s}$, where $L_{j}$ is the number of nodes in token ring $R_{j}$. We also define probabilistic token passing list, $\mathcal{L}(j)$, as the set of node IDs of all node members in token ring $R_{j}$. Each node in $R_{j}$ records a sequence of node IDs that the token has already visited for current token rotation cycle, and the current token holder selects the next token holder with equal probability from those nodes that have not been visited, to achieve fairness in channel access among all the nodes.

At the beginning of a superframe, a token starts to circulate among nodes in areas $A$ and $C$ during $T_{a c}$, forming token ring $R_{a c}$. Once a designated node in area $A(C)$ gets a token, it first waits for the channel to be idle for the duration of $T_{1}$ [20], and then piggybacks the token on the head-of-line (HOL) packet (if any) waiting in its queue and transmits the packet to its destination node in area $C(A)$. Note that the destination node (or the next-hop relay node) and the next token holder are not necessarily the same node. If the token holder does not have packets in its queue, it simply passes the token to the next token holder after $T_{1}$. When the current token rotation cycle finishes, a new token rotation cycle of token ring $R_{a c}$ starts, conforming to the same token passing rule until the end of $T_{a c}$. Once the duration $T_{a c}$ elapses, the current token circulation for $R_{a c}$ ceases, and another token starts to circulate among nodes in areas $C$ and $B$ during $T_{b c}$, forming token ring $R_{b c}$, which proceeds in the same way as in $T_{a c}$. The token rings $R_{a}$ and $R_{b}$ are formed when two tokens are circulated among nodes in area $A$ and among nodes in area $B$, respectively, during $T_{a b}$. These two token rings operate simultaneously and independently forming two disjoint one-hop subnetworks in the areas. Therefore, the durations, $T_{a c}, T_{b c}$, and $T_{a b}$, can be denoted by $k_{a c}, k_{b c}$, and $k_{a}\left(k_{b}\right)$ token rotation cycles for token rings, $R_{a c}$, $R_{b c}$, and $R_{a}\left(R_{b}\right)$, respectively, as shown in Fig. 4, indicating the number of times a token is held by each node in each token ring for packet transmissions. Note that the duration $T_{a b}$ can be denoted by $k_{a}$ token rotation cycles of $R_{a}$, or $k_{b}$ token rotation cycles of $R_{b}$ (not depicted in Fig. 4 for clarity). It is possible that the numbers of time slots, $M_{a c}, M_{b c}$, and $M_{a b}$, in each duration are not an integer multiple of the numbers of nodes, $L_{a c}, L_{b c}$, and $L_{a}\left(L_{b}\right)$, in respective token rings, making the numbers of token rotation cycles, $k_{j}(j=a c, b c, a, b)$, a noninteger. In this case, the number of time slots in the last token rotation cycle, denoted by $M_{j}-\left(\left\lceil k_{j}\right\rceil-1\right) L_{j}(\lceil\cdot\rceil$ is the ceiling function), is less than $L_{j}$. Since nodes are granted a random time slot in each token rotation cycle based on the probabilistic token passing, each node in token ring $R_{j}$ is statistically guaranteed to hold the token for $k_{j}$ times in each superframe. To ensure fair channel access among nodes, all the nodes in each token ring at least hold the token once in each duration (i.e., $k_{j} \geq 1$ ). Both $k_{j}$ and $M$ are MAC parameters that affect the performance of the TA-MAC scheme (to be discussed in Section III-E).

Any node, $x$, in the network, transmits two types of (data/token) packets: 1) type I and 2) type II packets. A type I packet contains a header, a set of IDs of the one-hop neighbors of node $x, \mathcal{N}(x)$, including the probabilistic token passing list,


Fig. 4. Token rotation cycles within $T_{a c}, T_{b c}$, and $T_{a b}$.
$\mathcal{L}(j)$, for the current token ring $R_{j}$, and a payload for either a data packet or a token packet. A type II packet is composed of a header and a payload. Each node in token ring $R_{j}$ ( $j=a c, b c, a, b$ ) transmits exactly one type I packet in the first token rotation cycle to exchange local information with its two-hop neighbors for detecting (updating) the node location and for distributedly calculating the durations, $T_{a c}, T_{b c}$, and $T_{a b}$, in each superframe. If more than one token rotation cycles are scheduled for token ring $R_{j}$, type II packets are transmitted in other token rotation cycles.

## B. Nodes Joining/Leaving the Network

A node needs to join corresponding token rings for packet transmissions when entering the network. To do so, it first specifies its location in the two-hop network. Suppose a new node, $x$, is powered on, and synchronizes in time with its onehop neighbors. Then, it listens to packet transmissions on the channel for one superframe duration, from which it obtains $\mathcal{N}(x)$. Then, the node determines the following.

1) It is an $S(D)$ node in areas $A$ or $B$, if $\exists \operatorname{ID}_{y} \in \mathcal{N}(x) \backslash \mathrm{ID}_{x}$, such that $\mathcal{N}(x) \subset \mathcal{N}(y)$, where $\mathrm{ID}_{x}$ and $\mathrm{ID}_{y}$ denote the IDs of node $x$ and node $y$.
2) It is an $R$ node in area $C$, if $\forall \mathrm{ID}_{y} \in \mathcal{N}(x) \backslash \mathrm{ID}_{x}$, we have $\mathcal{N}(y) \subseteq \mathcal{N}(x)$, and $\exists \mathrm{ID}_{z} \in \mathcal{N}(x) \backslash \mathrm{ID}_{x}$, such that $\mathcal{N}(z) \subset \mathcal{N}(x)$.
Furthermore, if node $x$ is an S-D node and can only receive packets from $R$ nodes in area $C$ during $T_{a c}\left(T_{b c}\right)$, it is located in area $B(A)$.

After determining its location, node $x$ broadcasts a REQUEST packet, which is a type III packet with a higher priority than types I and II data/token packets, after waiting for the channel to be idle for a duration of $T_{2}\left(<T_{1}\right)$, to join corresponding token rings. Each REQUEST packet contains a header, the transmitting node ID, and other two important information fields: 1) JOINING and 2) LEAVING, indicating the current network area that the node stays in and the previous area that it departed from. If the node is newly powered on, the LEAVING field is left blank. For instance, when node $x$ is powered on in area $A$, it broadcasts a REQUEST packet within $T_{a c}$, after sensing an idle channel for $T_{2}$, to join the token rings $R_{a c}$ and $R_{a}$, respectively. Upon receiving the REQUEST packet, each one-hop neighbor $y$ of $x$ adds $\mathrm{ID}_{x}$ in the set $\mathcal{N}(y)$ ( $\mathrm{ID}_{x}$ is added in the probabilistic token passing list in $\mathcal{N}(y)$ ). Consequently, if subsequent packet transmissions from any node $z$ in $R_{a c}\left(R_{a}\right)$ have $\mathrm{ID}_{x} \in \mathcal{N}(z)$ and $\mathrm{ID}_{x} \in \mathcal{L}(a c)(\mathcal{L}(a))$, the admissions to corresponding token rings are successful.

On the other hand, when node $x$ is expected to drain its power, it sends a REQUEST packet within $T_{a c}$ before
leaving area $A$, with the LEAVING field specifying area $A$ (the JOINING field is left blank). Then, each one-hop neighbor $y$ of $x$ removes $\mathrm{ID}_{x}$ from $\mathcal{N}(y)$. If node $x$ is the current token holder in $R_{a c}\left(R_{a}\right)$, the token is also passed to the next token holder after the REQUEST packet is transmitted before node departure.

## C. Existing Nodes Moving Across Network Areas

When an existing node, $x$, moves across network areas, its location change can be detected. ${ }^{3}$

1) When moving from areas $A$ to $C$, node $x$ detects its location change within $T_{b c}$ after receiving packets from nodes in area $B$. Then, it broadcasts a REQUEST packet, with JOINING and LEAVING fields specifying areas $C$ and $A$, to join $R_{b c}$ and leave $R_{a}$. Its ID, $\mathrm{ID}_{x}$, is added in $\mathcal{L}(b c)$ by each node in $R_{b c}$ and removed from $\mathcal{L}(a)$ by each node in $R_{a}$.
2) When moving from areas $C$ to $A$, node $x$ detects its location change within $T_{b c}$ if no packet transmission activity can be detected from nodes in area $B$. Similarly, a REQUEST packet is broadcast from node $x$ after the location change detection, and $\mathrm{ID}_{x}$ is then added in $\mathcal{L}(a)$ and removed from $\mathcal{L}(b c)$, respectively. For token ring $R_{b c}$, if the current token holder is a node from area $C$, it removes $\mathrm{ID}_{x}$ from $\mathcal{L}(b c)$ directly upon receiving the REQUEST packet. Any node, $y$, in area $B$ also removes $\mathrm{ID}_{x}$ from $\mathcal{N}(y)$ when receiving the updated $\mathcal{L}(b c)$ from the token holder. If the current token holder is a node from area $B$, it removes $\mathrm{ID}_{x}$ from $\mathcal{L}(b c)$ when it selects node $x$ as the next token holder and no transmission activity is detected within a retransmission timeout (see details in Section III-D).
Access collisions happen when two or more nodes, either newly arriving nodes or existing nodes, within the same onehop transmission range broadcast REQUEST packets at the same time, which can be detected by the nodes involved when their node IDs are not updated in corresponding token passing lists $\mathcal{L}(i)$ received from subsequent packet transmissions. Some random backoff-based collision resolution schemes can be used for the nodes involved before rebroadcasting a REQUEST packet [20]. When nodes move at a low speed (e.g., a walking speed), access collisions are unlikely to happen.

## D. Lost Token Recovery

Occasionally, existing nodes are not aware of a node departure in the following three situations. ${ }^{4}$

1) The REQUEST packet broadcast by a node being powered off is in collision, and new REQUEST packet cannot be reinitiated due to the power depletion.

[^3]2) The current token holder cannot correctly receive the broadcast REQUEST packet from a moving node since the communication range exceeds the one-hop distance.
3) The next token holder departs from the network due to node movement.
When one of the preceding situations happens, the token is lost, which can be detected by the previous token holder as there is no packet transmission from the current token holder. Then, the previous token holder enters into a token recovery process, in which it regenerates and passes a new token to the same current token holder for a maximum of $N_{r e}$ times [19], [22]. If there is still no transmission activity discovered after $N_{r e}$ is reached, a retransmission timeout is triggered and the previous token holder resends the token to a new node, with the old one removed from the probabilistic token passing list.

## E. Important MAC Parameters

The average packet service time for each node is the duration from the instant that a packet arrives at the head of a node queue to the instant it is successfully transmitted, averaged over all transmitted packets from the node. The average end-to-end delay in the two-hop network, denoted by $D_{a b}\left(D_{b a}\right)$, for the transmission direction from area $A(B)$ to $B(A)$ is the summation of the following.

1) The delay from the time a packet arrives at an $S$ node in area $A(B)$ to the time it is received by an $R$ node, averaged over all transmitted packets for the same transmission direction from area $A(B)$ to $C$, denoted by $D_{a c}\left(D_{b c}\right)$.
2) The delay from the moment a packet reaches a selected $R$ node to the moment it is received by its $D$ node in area $B(A)$, averaged over all transmitted packets for the same transmission direction from area $C$ to $B(A)$, denoted by $D_{c b}\left(D_{c a}\right)$.
To achieve minimal average end-to-end packet delay, each node in token ring $R_{j}(j=a c, b c, a, b)$ should determine the number of token rotation cycles $k_{j}(j=a c, b c, a, b)$ scheduled in $T_{a c}, T_{b c}$, and $T_{a b}$ of each superframe, which also indicates the number of time slots allocated to each node in token ring $R_{j}$. A small $k_{j}$ gives better time slot utilization in a token rotation cycle, i.e., the percentage of nonempty time slots in a token rotation cycle, due to the increased node queue utilization ratios, but can result in a longer transmission queue length for each node. Thus, with an increase of $k_{j}$, the packet delay is expected to be reduced due to more resource reservation for high service capability. However, excessive resource reservation for one token ring lowers the delay for single-hop packet transmissions within the token ring, but reduces the channel resources for other token rings, thus increasing the packet delay for other transmission hops. Moreover, an excessive $k_{j}$ prolongs the length of each superframe, which may cause the increase of average packet service time for each node. Therefore, to minimize the average end-to-end delay, first, time slot allocation for one individual token ring should be balanced with the others. Second, for the purpose of using a minimum total amount of channel resources to achieve the lowest average end-to-end delay, the total number of time slots
for each superframe, $M$, should also be optimized and adaptive to the varying numbers of nodes in each network area. We define $k_{j}^{\mathrm{opt}}, M^{\mathrm{opt}}$, and $D^{\mathrm{opt}}$ as the optimal number of token rotation cycles scheduled for token ring $R_{j}(j=a c, b c, a, b)$, the optimal total number of time slots for each superframe, and the minimal average end-to-end delay, respectively. Therefore, we aim at finding the set of MAC parameters $k_{j}^{\mathrm{opt}}$ and $M^{\mathrm{opt}}$, with a varying network load, to achieve $D^{\text {opt }}$ under the constraints that each node queue is stable and the average delay for local packet transmissions is bounded. With an increasing number of nodes in each network area, $M^{\text {opt }}$ is expected to consistently increase and then remain stable in high network load conditions to maintain a low packet service time, thus making $k_{j}^{\mathrm{opt}}$ decrease continuously. The set of optimal MAC parameters, $k_{j}^{\mathrm{opt}}$ and $M^{\mathrm{opt}}$, are distributedly calculated based on the current traffic load conditions for all the three network areas, and are also dynamically updated upon variations of the numbers of nodes in each area. According to $\left[k_{j}^{\mathrm{opt}}, M^{\mathrm{opt}}\right]$, nodes can also determine the optimal durations for $T_{a c}, T_{b c}$, and $T_{a b}$ in each superframe.

## IV. Performance Analysis

In this section, we develop performance analytical models for the TA-MAC scheme in closed-form functions of $k_{j}$ and $M$ for each superframe.

## A. Compound Packet Arrival Rate

Traffic arrivals at each node are modeled as a Poisson process with an arrival rate $\lambda$ packet/slot [20]. Each node in area $A$ $(B)$ transmits a packet either to a local destination node in the same area during $T_{a b}$, or to a relay node in area $C$ during $T_{a c}$ $\left(T_{b c}\right)$. Thus, the traffic arrivals at each node in area $A(B)$ are split into two streams with the average arrival rates denoted by $\lambda_{a}\left(\lambda_{b}\right)$ and $\lambda_{a c}\left(\lambda_{b c}\right)$ for each transmission direction. For analysis simplicity, we assume that every packet generated from an $S$ node's own application layer is transmitted equally likely for both directions. Thus, $\lambda_{a}\left(\lambda_{b}\right)$ and $\lambda_{a c}\left(\lambda_{b c}\right)$ are equal to ( $\lambda / 2$ ) packet/slot. On the other hand, traffic arrivals at each relay node consist of two portions.

1) Traffic generated at the node's own application layer and destined for a node either in areas $A$ or $B$ equally as assumed for an $S$ node, with the average arrival rate ( $\lambda / 2$ ) packet/slot for both transmission directions.
2) The relay traffic received from nodes in area $A(B)$ during $T_{a c}\left(T_{b c}\right)$, which will be forwarded to a destination node chosen in area $B(A)$ during $T_{b c}\left(T_{a c}\right)$.
To analyze the average end-to-end delay from a source node in area $A(B)$ to a destination node in area $B(A)$ as well as the aggregate throughput for the two-hop network, a network of queues should be established. However, the exact relay traffic arrival process at each node in area $C$, consisting of the superposition of the departure processes from the nodes in area $A(B)$, is difficult to be precisely modeled [26]. Therefore, inspired by Omar et al. [13], we approximate the relay traffic arrival process at a relay node as a single Poisson process with rate parameter, $\lambda_{a r}\left(\lambda_{b r}\right)$, being the sum of the traffic arrival rates heading to the common relay node from nodes


Fig. 5. Poisson approximation for relay traffic arrival rate for transmission direction from areas $C$ to $B$.
in area $A(B)$, as shown in Fig. 5. Under the assumption that each source node selects its relay from $N_{c}$ nodes with equal probability, the compound traffic arrival rates at each relay node with the combination of the external traffic and the relay traffic for the transmission directions from areas $C$ to $B$ and from areas $C$ to $A$, denoted by $\lambda_{c b}$ and $\lambda_{c a}$, respectively, are approximately given by

$$
\begin{align*}
& \lambda_{c b} \approx \frac{\lambda}{2}+\lambda_{a r}=\frac{\lambda}{2}+\frac{\lambda N_{a}}{2 N_{c}} \quad(C \rightarrow B)  \tag{1}\\
& \lambda_{c a} \approx \frac{\lambda}{2}+\lambda_{b r}=\frac{\lambda}{2}+\frac{\lambda N_{b}}{2 N_{c}} \quad(C \rightarrow A) \tag{2}
\end{align*}
$$

As an extension to Kleinrock independence approximation [26], this Poisson traffic approximation on each relay node is effective in analytically modeling the two-hop network as a network of M/G/1 queues for evaluating the average end-to-end delay. To justify the accuracy of this approximation, the analytical results are further compared with the simulation results in Section VI-B. The approximation becomes more accurate when the network traffic load increases [26] (i.e., a large number of nodes in each network area with increased queue utilization ratios for each node), under which the TA-MAC scheme can also achieve high channel utilization.

## B. Average Packet Service Time

We calculate average packet service time for HOL packet transmissions. Packet service time (in the unit of time slot), $W_{s, j}$, for a node in token ring $R_{j}(j=a c, b c, a, b)$, is the duration from the instant that a packet arrives at the head of the node queue to the instant it is successfully transmitted. We use index $q$ ranging from 1 to $L_{j}$ to indicate the end instant of each time slot in a token rotation cycle of $R_{j}$. Take nodes in token ring $R_{a c}$ as an example, in which there are $k_{a c}$ token rotation cycles scheduled within each $T_{a c}$. HOL packets from a tagged node $x$ in areas $A$ and $C$ can appear at the end of each allocated time slot in any of $k_{a c}$ token rotation cycles along each $T_{a c}$. For analysis tractability, we neglect the possibility that an HOL packet arrives within the duration of a time slot, and the possibility that a packet arrives at a node in $R_{a c}$ and finds the node queue empty during $T_{b c}$ and $T_{a b}$ [27], which is more unlikely for a higher traffic load condition. We make the same assumption for HOL packet arrivals at nodes in all token rings. Define a random variable $H_{j}$, which equals 0 if an HOL packet from a tagged node in $R_{j}$ appears during the $k_{j}$ th token rotation cycle, and equals 1 otherwise. As each node has a randomly
selected transmission slot in the corresponding token passing sequence of each token rotation cycle, the service times for consecutive packet transmissions are independent and identically distributed (i.i.d) random variables [13], which is a necessary condition for the $\mathrm{M} / \mathrm{G} / 1$ queue modeling for each node. Thus, the HOL packet is transmitted in a time slot randomly chosen from the next token rotation cycle following the packet arriving instant. Therefore, we derive the average packet service time, denoted by $E\left[W_{s, j}\right]$, by considering the following two cases.

1) When $H_{j}=1$, the probability mass function (pmf) of $W_{s, j}$ conditioned on the arriving time instant $Q$ of the HOL packet is expressed as

$$
\begin{align*}
& P\left\{W_{s, j}=i \mid Q=q, H_{j}=1\right\} \\
& \quad=\frac{1}{L_{j}}\left(L_{j}-q+1 \leq i \leq 2 L_{j}-q, 1 \leq q \leq L_{j}\right) \tag{3}
\end{align*}
$$

where $L_{a c}=N_{a}+N_{c}, L_{b c}=N_{b}+N_{c}, L_{a}=N_{a}, L_{b}=N_{b}$, and the expectation of $W_{s, j}$ conditioned on $H_{j}=1$ can be derived as

$$
\begin{align*}
E\left[W_{s, j} \mid H_{j}=1\right]= & \sum_{i=L_{j}-q+1}^{2 L_{j}-q} \sum_{q=1}^{L_{j}} i P\left\{W_{s, j}=i \mid Q=q, H_{j}=1\right\} \\
& \times P\left\{Q=q \mid H_{j}=1\right\} \\
= & L_{j} \tag{4}
\end{align*}
$$

2) When $H_{j}=0$, we similarly have the conditional pmf and the conditional expectation of $W_{s, j}$ as

$$
\begin{array}{r}
P\left\{W_{s, j}=i \mid Q=q, H_{j}=0\right\}=\frac{1}{L_{j}}\left(U_{j}-q+1 \leq i \leq U_{j}\right. \\
\left.-q+L_{j}, 1 \leq q \leq L_{j}\right) \tag{5}
\end{array}
$$

where $U_{j}=M-\left(k_{j}-1\right) L_{j}$, and

$$
\begin{align*}
E\left[W_{s, j} \mid H_{j}=0\right]= & \sum_{i=U_{j}-q+1}^{U_{j}-q+L_{j}} \sum_{q=1}^{L_{j}} i P\left\{W_{s, j}=i \mid Q=q, H_{j}=0\right\} \\
& \times P\left\{Q=q \mid H_{j}=0\right\} \\
= & U_{j} \tag{6}
\end{align*}
$$

Therefore, the average service time for each HOL packet from nodes in $R_{j}$ can be derived as

$$
\begin{align*}
E\left[W_{s, j}\right]= & E\left[W_{s, j} \mid H_{j}=1\right] P\left\{H_{j}=1\right\} \\
& +E\left[W_{s, j} \mid H_{j}=0\right] P\left\{H_{j}=0\right\} \\
= & L_{j} \cdot\left(1-\frac{1}{k_{j}}\right)+U_{j} \cdot \frac{1}{k_{j}} \\
= & \frac{M}{k_{j}}, j=a c, b c, a, b \tag{7}
\end{align*}
$$

Define the average packet service rate, $\mu_{j}$, as the number of packets transmitted per slot time from a node in $R_{j}$. Thus

$$
\begin{equation*}
\mu_{j}=\frac{1}{E\left[W_{s, j}\right]}=\frac{k_{j}}{M}, \quad j=a c, b c, a, b \tag{8}
\end{equation*}
$$

## C. Aggregate Network Throughput

The aggregate network throughput is defined as the ratio of average number of transmitted packets over total number of time slots in each superframe, which is also the aggregate time slot utilization for each superframe, given by

$$
\begin{align*}
S=\frac{1}{M}( & k_{a c} \frac{N_{a} \lambda_{a c}+N_{c} \lambda_{c a}}{\mu_{a c}} \\
& \left.+k_{b c} \frac{N_{b} \lambda_{b c}+N_{c} \lambda_{c b}}{\mu_{b c}}+\sum_{j \in\{a, b\}} k_{j} \frac{N_{j} \lambda_{j}}{\mu_{j}}\right) . \tag{9}
\end{align*}
$$

From (7) and (9), the aggregate network throughput, $S$, is a function of the numbers of nodes in each network area and the traffic arrival rates at the nodes. Actually, given certain numbers of nodes in the network areas, the variations of $k_{j}$ and $M$ affect the average packet service rate $\mu_{j}$ for each node in $R_{j}$ and the time slot utilization for each token rotation cycle, resulting in a different packet delay. But the aggregate channel utilization for an entire superframe remains unchanged with variations of $k_{j}$ and $M$. Therefore, higher aggregate network throughput is expected to be achieved with more nodes in all individual network areas.

## D. Average End-to-End Delay

As defined in Section III-E, the average end-to-end delay in the two-hop network, $D_{a b}\left(D_{b a}\right)$, for the transmission direction from areas $A(B)$ to $B(A)$ is the summation of average packet delays $D_{a c}\left(D_{b c}\right)$ and $D_{c b}\left(D_{c a}\right)$. The average packet delay is composed of the average queueing delay, i.e., the average duration the packet stays in the transmission queue after its arrival, and the average service time.

We derive the second moment of the packet service time $W_{s, j}$ for each node in token ring $R_{j}(j=a c, b c, a, b)$ as

$$
\begin{align*}
E\left[W_{s, j}^{2}\right]= & E\left[W_{s, j}^{2} \mid H_{j}=1\right] P\left\{H_{j}=1\right\} \\
& +E\left[W_{s, j}^{2} \mid H_{j}=0\right] P\left\{H_{j}=0\right\} \tag{10}
\end{align*}
$$

Then, based on $P$ - $K$ formula [26], the average end-to-end delay, $D_{a b}\left(D_{b a}\right)$, for either transmission direction, and the average delay for local transmissions within area $A(B)$, denoted by $D_{a}\left(D_{b}\right)$, can be derived as

$$
\begin{align*}
D_{a b} & =D_{a c}+D_{c b} \\
& =\sum_{(n, j) \in\{(a c, a c),(c b, b c)\}}\left(E\left[W_{s, j}\right]+\frac{\lambda_{n} E\left[W_{s, j}^{2}\right]}{2\left(1-\lambda_{n} E\left[W_{s, j}\right]\right)}\right) \\
& =\sum_{(n, j) \in\{(a c, a c),(c b, b c)\}}\left(\frac{\varepsilon_{j}}{k_{j}}+\frac{\lambda_{n}\left[\alpha_{j} k_{j}^{2}+\beta_{j} k_{j}+\gamma_{j}\right]}{2\left(k_{j}-\lambda_{n} \varepsilon_{j}\right)}\right)  \tag{11}\\
D_{b a} & =D_{b c}+D_{c a} \\
& =\sum_{(n, j) \in\{(b c, b c),(c a, a c)\}}\left(\frac{\varepsilon_{j}}{k_{j}}+\frac{\lambda_{n}\left[\alpha_{j} k_{j}^{2}+\beta_{j} k_{j}+\gamma_{j}\right]}{2\left(k_{j}-\lambda_{n} \varepsilon_{j}\right)}\right) \tag{12}
\end{align*}
$$

and

$$
\begin{equation*}
D_{j}=\frac{\varepsilon_{j}}{k_{j}}+\frac{\lambda_{j}\left[\alpha_{j} k_{j}^{2}+\beta_{j} k_{j}+\gamma_{j}\right]}{2\left(k_{j}-\lambda_{j} \varepsilon_{j}\right)}, \quad j=a, b \tag{13}
\end{equation*}
$$

where $\alpha_{j}=L_{j}^{2}, \beta_{j}=-\left(5 L_{j}^{2}+12 M L_{j}+1 / 6\right), \gamma_{j}=M^{2}+2 M L_{j}$, and $\varepsilon_{j}=M$.

From (11)-(13), it is observed that, with certain numbers of nodes in each network area, both average end-to-end delay and average delay for local transmissions are functions of $k_{j}$ and $M$. As stated in Section III-E, with a certain $M$ value, an increased $k_{j}$ for one token ring reduces the one-hop average packet delay among its node members, and also the time resources for other token rings, which can increase the average delay for other transmission hops. Thus, the numbers of token rotation cycles scheduled for each token ring should be balanced to minimize the average end-to-end delay. At the same time, the total number of time slots, $M$, for each superframe should be properly chosen to further improve the delay performance. Therefore, our design objective is to determine $k_{j}$ and $M$ to achieve the minimal average end-to-end delay.

## V. Optimal MAC Parameters

In this section, we propose an optimization framework to find the set of optimal MAC parameters, $\left[k_{j}^{\mathrm{opt}}, M^{\mathrm{opt}}\right]$, for each superframe, with which the minimal average end-to-end delay, $D^{\text {opt }}$, is achieved, and a bounded average delay, $D_{\text {th }}$, for local transmissions is guaranteed.

## A. Average End-to-End Delay Minimization

An average end-to-end delay minimization problem is formulated as $(P 1)$, to derive the set of optimal numbers of token rotation cycles, $\mathbf{k}^{*}=\left[k_{a c}^{*}, k_{b c}^{*}, k_{a}^{*}, k_{b}^{*}\right]$, for each token ring, for a given $M$

$$
\begin{align*}
& (P 1): \min _{\mathbf{k}=\left[k_{a c}, k_{b c}, k_{a}, k_{b}\right]}\left\{\max \left\{D_{a b}\left(k_{a c}, k_{b c}\right), D_{b a}\left(k_{a c}, k_{b c}\right)\right\}\right\} \\
& \text { s.t. }\left\{\begin{array}{l}
k_{a c} L_{a c}+k_{b c} L_{b c}+k_{a} L_{a}=M \\
k_{a} L_{a}=k_{b} L_{b} \\
\rho_{n}=\frac{\lambda_{n}}{\mu_{j}}<1, \quad(n, j) \in\{(c a, a c),(c b, b c)\} \\
\rho_{j}=\frac{\lambda_{j}}{\mu_{j}}<1, \quad j=a, b \\
D_{j}\left(k_{j}\right) \leq D_{\mathrm{th}}, \quad j=a, b \\
k_{j} \geq 1, \quad j=a c, b c, a, b .
\end{array}\right. \tag{14a}
\end{align*}
$$

In $(P 1)$, the objective is to minimize the average end-to-end delay for both transmission directions from areas $A$ to $B$ and from areas $B$ to $A$, by finding the set of optimal numbers of token rotation cycles for each token ring in each superframe. Constraints (14a) and (14b) indicate the total number of time slots for each superframe is $M$, and the time slots allocated to nodes in areas $A$ and $B$ are balanced based on the numbers of nodes in both areas. Constraint (14c) guarantees a stable relay node queue in token rings $R_{a c}$ and $R_{b c}$, where $\rho_{n}=\left(\lambda_{n} / \mu_{j}\right)$
denotes queue utilization ratio for relay nodes in area $C .{ }^{5}$ Similarly, $\rho_{j}$ in constraint (14d) denotes queue utilization ratio for nodes in token rings $R_{a}$ and $R_{b}$. Constraint (14e) states that the average delays for local packet transmissions within both areas $A$ and $B$ are bounded by threshold $D_{\text {th }}$. Constraint (14f) guarantees that nodes in each token ring hold the token at least once in each superframe to ensure fair channel access.

The main difficulty to solve ( $P 1$ ) is the nonconvexity of the objective function in terms of $\mathbf{k}$. Therefore, to make the problem tractable, we decouple ( $P 1$ ) into a convex subproblem and a biconvex subproblem [28], [29] with two separate sets of decision variables. By solving these two subproblems sequentially, the original problem can be solved. First, we introduce an important proposition and its corollary.

Proposition 1: In (P1), the one-hop average delays, $D_{a c}$ $\left(D_{c a}\right), D_{b c}\left(D_{c b}\right)$, and $D_{a}\left(D_{b}\right)$, for packet transmissions from areas $A(C)$ to $C(A)$, from areas $B(C)$ to $C(B)$, and within the local area $A(B)$, are all strictly convex functions in terms of $k_{a c}, k_{b c}$, and $k_{a}\left(k_{b}\right)$, respectively.

The proof of Proposition 1 is given in Appendix A.
Corollary 1: In ( $P 1$ ), the one-hop average delays, $D_{a c}$ $\left(D_{c a}\right), D_{b c}\left(D_{c b}\right)$, and $D_{a}\left(D_{b}\right)$, are strictly decreasing functions of $k_{a c}, k_{b c}$, and $k_{a}\left(k_{b}\right)$, respectively.
The proof of Corollary 1 is given in Appendix B.
According to Corollary 1, each one-hop average delay is a decreasing function of $k_{j}$ for corresponding token ring. Thus, to minimize the average end-to-end delay, the maximum amount of time slots are expected to be allocated to nodes in token rings $R_{a c}$ and $R_{b c}$ among all feasible solutions for ( $P 1$ ). In other words, the minimum number of time slots should be reserved for the token rings $R_{a}$ and $R_{b}$ under constraints (14b)-(14f). Therefore, to tackle (P1) efficiently, we first solve the following subproblem (SP1) to obtain the set of optimal numbers of token rotation cycles, $\mathbf{k}_{\mathbf{1}}^{*}=\left[k_{a}^{*}, k_{b}^{*}\right]$, scheduled for token rings $R_{a}$ and $R_{b}$, respectively,

$$
\begin{align*}
& (\mathrm{SP} 1): \min _{\mathbf{k}_{\mathbf{1}}=\left[k_{a}, k_{b}\right]} k_{a} L_{a} \\
& \text { s.t. }\left\{\begin{array}{l}
k_{a} L_{a}=k_{b} L_{b} \\
\rho_{j}<1, \quad j=a, b \\
D_{j}\left(k_{j}\right) \leq D_{\mathrm{th}}, \quad j=a, b \\
k_{j} \geq 1, \quad j=a, b
\end{array}\right. \tag{15a}
\end{align*}
$$

We further simplify (SP1) as (SP1a) with a single decision variable $k_{a}$, by substituting constraint (15a) into constraints (15b)-(15d)

$$
\begin{align*}
& \text { (SP1a) }: \min _{k_{a}} k_{a} L_{a} \\
& \text { s.t. }\left\{\begin{array}{l}
\rho_{a}<1 \\
\rho_{b}\left(\frac{k_{a} L_{a}}{L_{b}}\right)<1 \\
D_{a}\left(k_{a}\right) \leq D_{\mathrm{th}} \\
D_{b}\left(\frac{k_{a} L_{a}}{L_{b}}\right) \leq D_{\mathrm{th}} \\
\\
\frac{\max \left\{1, \frac{L_{b}}{L_{a}}\right\}}{k_{a}} \leq 1
\end{array}\right. \tag{16a}
\end{align*}
$$

[^4]In (SP1a), the objective function and the left-hand sides of all inequality constraints (16a)-(16c) and (16e) are convex functions of $k_{a}$. For constraint (16d), the function $D_{b}\left(k_{a} L_{a} / L_{b}\right)$ is a composite function of $k_{a}$, where $D_{b}(\cdot)$ is a convex and decreasing function of $k_{b}$ and $k_{b}=\left(k_{a} L_{a} / L_{b}\right)$ is a linear function of $k_{a}$. Therefore, according to the scalar composition rules [30], $D_{b}\left(k_{a} L_{a} / L_{b}\right)$ is also a convex function of $k_{a}$. Hence, (SP1a) is proved to be a convex optimization problem, which can be efficiently solved to get the optimal solution $k_{a}^{\dagger}$. Since the total number of time slots, $M_{a b}$, reserved for $R_{a}$ or $R_{b}$ is required to be integer, we obtain the optimal numbers of token rotation cycles for each of the token rings as

$$
\begin{equation*}
k_{a}^{*}=\frac{\left\lceil k_{a}^{\dagger} L_{a}\right\rceil}{L_{a}} \tag{17}
\end{equation*}
$$

and

$$
\begin{equation*}
k_{b}^{*}=\frac{\left\lceil k_{a}^{\dagger} L_{a}\right\rceil}{L_{b}} \tag{18}
\end{equation*}
$$

Note that $k_{a}^{*}$ and $k_{b}^{*}$ are guaranteed the global optimal in the feasible set of (SP1a), since all the inequality constraint functions of (SP1a) are decreasing functions of the decision variable $k_{a}$.

By substituting the optimal set of values $\left[k_{a}^{*}, k_{b}^{*}\right]$ into $(P 1)$, the original optimization problem is reduced to the second subproblem (SP2)

$$
\begin{gather*}
(\mathrm{SP} 2): \min _{\mathbf{k}_{\mathbf{2}}=\left[k_{a c}, k_{b c}\right]}\left\{\max \left\{D_{a b}\left(k_{a c}, k_{b c}\right), D_{b a}\left(k_{a c}, k_{b c}\right)\right\}\right\} \\
\text { s.t. }\left\{\begin{array}{l}
k_{a c} L_{a c}+k_{b c} L_{b c}=M^{*} \\
\rho_{n}<1, \quad n=c a, c b \\
k_{j} \geq 1, \quad j=a c, b c
\end{array}\right. \tag{19a}
\end{gather*}
$$

where $M^{*}=M-k_{a}^{*} L_{a}$.
Theorem 1: In (SP2), the 2-D decision vector $\mathbf{k}_{2}$ represents a biconvex set, if $\mathbf{k}_{\mathbf{2}}$ is a convex set with respect to $k_{b c}\left(k_{a c}\right)$ for any given $k_{a c}\left(k_{b c}\right)$ from the feasible solutions.

Proof: Rewrite the set of constraints of (SP2) in a standard form for a given $k_{a c}\left(k_{b c}\right)$ in $\mathbf{k}_{\mathbf{2}}$, the equality constraint is an affine function of $k_{b c}\left(k_{a c}\right)$, and both inequality constraints are convex functions of $k_{b c}\left(k_{a c}\right)$. Therefore, the set of feasible solutions of $k_{b c}\left(k_{a c}\right)$ satisfying all the constraints form a convex set [30].

Theorem 2: In (SP2), the objective function defined on the biconvex set $\mathbf{k}_{2}$ represents a biconvex function, if the objective function is a convex function in terms of $k_{b c}\left(k_{a c}\right)$ for any given $k_{a c}\left(k_{b c}\right)$ from the feasible solutions.

Proof: Given $k_{a c}\left(k_{b c}\right)$ in $\mathbf{k}_{\mathbf{2}}$, both $D_{a b}(\cdot)$ and $n_{b a}(\cdot)$ functions are a linear combination of a convex function in terms of $k_{b c}\left(k_{a c}\right)$ and a constant, which are also convex. Moreover, the max function, $\max \{x, y\}$, proved to be convex on $\mathbf{R}^{2}$ in [30], is also nondecreasing in each of its two arguments. Therefore, according to the vector composition rules, [30], the objective function $\max \left\{D_{a b}, D_{b a}\right\}$ is a convex function with respect to $k_{b c}\left(k_{a c}\right)$.

Based on Theorems 1 and 2, (SP2) is a biconvex optimization problem since we have a biconvex objective function minimized over a biconvex set, which often has multiple local optima and is difficult to determine the global optimal solution [31]. Therefore, to solve (SP2) efficiently, we
further simplify (SP2) into a single variable optimization problem (SP2a), by substituting equality constraint (19a) into the objective function and other constraints

$$
\begin{align*}
&(\mathrm{SP} 2 \mathrm{a}): \min _{k_{a c}}\left\{\max \left\{D_{a b}\left(k_{a c}, h\left(k_{a c}\right)\right), D_{b a}\left(k_{a c}, h\left(k_{a c}\right)\right)\right\}\right\} \\
& \text { s.t. }\left\{\begin{array}{l}
\rho_{c a}<1 \\
\rho_{c b}\left(h\left(k_{a c}\right)\right)<1 \\
1 \leq k_{a c} \leq \frac{M^{*}-L_{b c}}{L_{a c}}
\end{array}\right. \tag{20a}
\end{align*}
$$

where $h\left(k_{a c}\right)=k_{b c}=\left(M^{*}-k_{a c} L_{a c} / L_{b c}\right)$.
Proposition 2: (SP2a) is a convex optimization problem, with respect to the decision variable $k_{a c}$.

The proof of Proposition 2 is given in Appendix C.
Based on Proposition 2, the convex optimization problem (SP2a) can also be efficiently solved for the optimal solutions $k_{a c}^{\dagger}$ and $k_{b c}^{\dagger}$. Similar to (17) and (18), we further obtain the optimal numbers of token rotation cycles, $k_{a c}^{*}$ and $k_{b c}^{*}$, for $R_{a c}$ and $R_{b c}$, respectively, by rounding $k_{a c}^{\dagger} L_{a c}$ to the integer number (within the feasible region) that achieves the minimum value of the objective function in (SP2a).

Based on the optimal numbers of token rotation cycles, $\mathbf{k}^{*}=$ $\left[k_{a c}^{*}, k_{b c}^{*}, k_{a}^{*}, k_{b}^{*}\right]$, in each superframe under a predefined $M$, the average end-to-end packet delay is minimized, denoted as $D^{*}$.

## B. Optimal Total Number of Time Slots for Each Superframe

As discussed in Section III-E, $D^{*}$ decreases with an increase of $M$, due to more time slots reserved for each token ring. If $M$ is set too large, excessive resource reservation prolongs the superframe length, causing an increase of packet delay. Therefore, we aim at determining the optimal total number of time slots, $M^{\text {opt }}$, for each superframe, given the optimal number of token rotation cycles, $k_{j}^{\mathrm{opt}}$, for token ring $R_{j}(j=a c, b c, a, b)$, to minimize average end-to-end delay. To this end, we propose an optimal superframe length calculation algorithm, Algorithm 1, for each node to distributedly determine and update the set of optimal MAC parameters, $\left[k_{j}^{\mathrm{opt}}, M^{\mathrm{opt}}\right]$. The procedure of the algorithm is summarized as follows.

Step 1: The minimum value for $M$ is set to satisfy constraint (14f) in (P1). Both average end-to-end delay and $M_{s}$ [the minimum value of $M$ to make ( $P 1$ ) feasible] are initialized, and the maximum iteration limit is set to a large number.

Step 2: The sequential subproblems (SP1a) and (SP2a) are repeatedly solved by increasing $M$ with the increment of one time slot in each iteration until a set of feasible solutions, $\left[k_{j}^{*}, D^{*}\right]$, for $(P 1)$ are found at $M=M_{s}$ (if no feasible $M_{s}$ is found at the maximum iteration limit under current network load condition, the newly arriving node will not be admitted so that the numbers of nodes in each network area can be controlled within the network capacity).

Step 3: Starting from a feasible $M_{s}$, we iteratively search for $M^{\mathrm{opt}}$ and $k_{j}^{\mathrm{opt}}$ to achieve the minimal average end-to-end delay $D^{\text {opt }}$, by continuously increasing $M$ and solving (SP1a) and (SP2a) at each updated $M$ until the maximum iteration limit is reached.

```
Algorithm 1: Optimal Superframe Length Calculation
Algorithm
    Input : \(L_{a c}, L_{b c}, L_{a}, L_{b}, \lambda, D_{\mathrm{th}}\).
    Output: \(k_{j}^{o p t}, D^{o p t}, M^{o p t}\), and \(T_{f}^{o p t}\).
    Initialization: \(M \leftarrow L_{a c}+L_{b c}+\max \left\{L_{a}, L_{b}\right\}\),
    \(D^{o p t} \leftarrow+\infty, M_{s} \leftarrow 0\), set the maximum iteration limit;
    do
        \(\left[k_{j}^{*}, D^{*}\right] \leftarrow\) solving (SP1a) and (SP2a);
        if No feasible solutions are found then
            if The maximum iteration limit is reached then
                break;
            end
            \(M \leftarrow M+1 ;\)
        else
            \(M_{s} \leftarrow M ;\)
            break;
        end
    while (P1) is not feasible;
    if \(M_{s}>0\) then
        while The maximum iteration limit is not reached do
            if \(D^{*}<D^{o p t}\) then
                \(k_{j}^{\text {opt }} \leftarrow k_{j}^{*} ;\)
                \(D^{o p t} \leftarrow D^{*}\);
                \(M^{o p t} \leftarrow M ;\)
            end
            \(M \leftarrow M+1 ;\)
            \(\left[k_{j}^{*}, D^{*}\right] \leftarrow\) solving (SP1a) and (SP2a);
            end
            \(T_{f}^{o p t} \leftarrow M^{o p t} \cdot T_{s} ;\)
            return \(k_{j}^{o p t}, M^{o p t}, T_{f}^{o p t}\), and \(D^{o p t}\).
    end
```


## VI. Numerical Results

In this section, analytical and simulation results are presented to demonstrate the accuracy of performance analysis. All the simulations are carried out using OMNeT++ [32], [33]. In the simulation, nodes are scattered over a $150 \times 150 \mathrm{~m}$ square region, forming a two-hop network with three network areas similar to that shown in Fig. 1(b), where nodes within the transmission range (set to 50 m ) of all other nodes can relay traffic from the S-D node pairs that are not reachable to each other directly. For each transmitted packet, the source node randomly selects a next-hop node or a destination node, according to the packet's destination area, among a specific group of nodes. External traffic arrivals for each node are modeled as a Poisson process with the rate of 0.01 packet/slot ( 10 packet/s). The delay bound for local packet transmissions within areas $A$ and $B$ is set as 400 ms . Each simulation point is generated by running the simulation for 10000 superframes. The main simulation parameters are summarized in Table II.

By solving the sequential subproblems (SP1a) and (SP2a) using Algorithm 1, we determine the optimal total number of time slots, $M^{\text {opt }}$, for each superframe, and the optimal number

TABLE II
Simulation Parameters

| Parameters MAC schemes | TA-MAC | LA-MAC [16] | DTSA [15] [16] |
| :--- | :---: | :---: | :---: |
| Channel capacity | 11 Mbps | 11 Mbps | 11 Mbps |
| Time slot duration | 1 ms | 1 ms | 1 ms |
| Idle duration $\left(T_{1} / T_{2}\right)$ | $20 / 10 \mu \mathrm{~s}$ | $\mathrm{n} . \mathrm{a}$. | n.a. |
| Node ID | 7 bit | 7 bit | 7 bit |
| Data/Token packet duration | $0.98 / 0.28 \mathrm{~ms}$ | $0.98 \mathrm{~ms} / \mathrm{n} . \mathrm{a}$. | $0.98 \mathrm{~ms} / \mathrm{n} . \mathrm{a}$. |
| REQUEST packet duration | 0.22 ms | $\mathrm{n} . \mathrm{a}$. | $\mathrm{n} . \mathrm{a}$. |
| Packet arrival rate $(\lambda)$ | $0.01 \mathrm{packet} / \mathrm{slot}$ | $0.01 \mathrm{packet} / \mathrm{slot}$ | $0.01 \mathrm{packet} / \mathrm{slot}$ |
| Transmission queue length | 10000 packets | 10000 packets | 10000 packets |
| Delay bound for local transmissions $\left(D_{t h}\right)$ | 400 ms | $\mathrm{n} . \mathrm{a}$. | n.a. |



Fig. 6. Optimal number of token rotation cycles $k_{j}^{*}$ for token ring $R_{j}(j=$ $a c, b c, a, b)$ versus $M\left(N_{a}=20, N_{b}=15, N_{c}=15\right)$.
of token rotation cycles, $k_{j}^{\mathrm{opt}}$, scheduled for token ring $R_{j}(j=$ $a c, b c, a, b$ ), upon which the minimal average end-to-end delay $D^{\text {opt }}$, the bounded average delays for local transmissions, $D_{a}$ and $D_{b}$, in areas $A$ and $B$, can be achieved by the TA-MAC scheme. Then, we analyze $D^{\mathrm{opt}}, D_{a}$, and $D_{b}$, and the aggregate network throughput with respect to variations of the network traffic load. Lastly, the TA-MAC scheme is compared with a hybrid MAC scheme and a dynamic TDMA scheme in terms of delay and throughput over a wide range of network traffic load.

## A. Optimal MAC Parameters

Fig. 6 shows the optimal number of token rotation cycles, $k_{j}^{*}$, scheduled for token ring $R_{j}(j=a c, b c, a, b)$ versus $M$. We can see that $k_{j}^{*}$ increases with $M$, and more token rotation cycles are scheduled for token rings $R_{a c}$ and $R_{b c}$ to minimize the average end-to-end delay. The set of optimal MAC parameters, $\left[M^{\mathrm{opt}}, k_{j}^{\mathrm{opt}}\right]$, is also shown, based on Algorithm 1, which achieves the minimal average end-to-end delay.
Fig. 7 demonstrates the minimal average end-to-end delay, $D^{*}$, and the average delays for local packet transmissions, $D_{a}$ and $D_{b}$, versus $M$. It can be seen that $D^{*}$ decreases with $M$ when $M$ is small, which indicates that more token rotation cycles are required to achieve a smaller end-to-end delay. When $M$ becomes large, $D^{*}$ starts to increase since excessive time slot reservation for each superframe increases packet service time. Therefore, the optimal total number of time slots for each superframe, $M^{\text {opt }}$, can be determined based on Algorithm 1 to achieve the minimal average end-to-end


Fig. 7. Average packet delay versus $M\left(N_{a}=20, N_{b}=15, N_{c}=15\right)$.


Fig. 8. Optimal total number of time slots, $M^{\mathrm{opt}}$, for each superframe and the optimal number of token rotation cycles, $k_{j}^{\mathrm{opt}}$, for token ring $R_{j}$ with respect to the total number of nodes, $N\left(N_{a}=N_{b}=N_{c}\right)$.
delay, $D^{\text {opt }}$. We can also see that the average delays for local transmissions, $D_{a}$ and $D_{b}$, are below threshold $D_{\mathrm{th}}=400 \mathrm{~ms}$. Nodes in token rings $R_{a}$ and $R_{b}$ are always guaranteed the minimum amount of time slots to maintain bounded average delays for local packet transmissions.

In Fig. 8 , we evaluate $M^{\mathrm{opt}}$ and $k_{j}^{\mathrm{opt}}$ as the total number of nodes, $N$, in the network increases, with the same numbers of nodes in each area ( $N_{a}=N_{b}=N_{c}$ ). It can be seen that $M^{\text {opt }}$ increases consistently with $N$ and remains at a steady value when the network load becomes high, which indicates that the optimal superframe length for the TA-MAC is adaptive to the network traffic load variations and is stable at high traffic load conditions. Within each superframe, the optimal numbers


Fig. 9. Evaluation of (a) average delay for relay transmissions and (b) average end-to-end delay, under different network load conditions.

(a)

(b)

Fig. 10. Evaluation of average delay for local transmissions (a) in area A and (b) in area B, under different network load conditions.
of token rotation cycles, $k_{a c}^{\mathrm{opt}}$ and $k_{b c}^{\mathrm{opt}}$, keep decreasing with an increase of $N$, and $k_{a}^{\mathrm{opt}}$ and $k_{b}^{\mathrm{opt}}{ }^{m c}$ maintain at the minimum value in order to provide the maximum amount of resources for nodes in $R_{a c}$ and $R_{b c}$ to achieve the minimal average end-to-end delay.

## B. Performance Metrics for the TA-MAC

We evaluate the average delay for relay packet transmissions and the average end-to-end delay for both low and high network traffic load conditions, as the number of nodes, $N_{a}$, in area $A$ varies. In Fig. 9(a), it is shown that the average delay for relay transmissions increases consistently with $N_{a}$, and the simulation results match the analytical results more closely when the network traffic load becomes high ( $N_{c}=25, N_{b}=15$ ), which verifies the effectiveness of the Poisson compound traffic arrival rate approximation on each relay node used in the analysis. Basically, the approximation becomes more accurate with an increasing number of nodes and node queue utilization ratios. Similar trends are observed in Fig. 9(b). As expected, the minimal average end-to-end delay increases with the number of nodes. The simulation results match well with the analytical results.

Fig. 10(a) and (b) demonstrate average delays for local transmissions in areas $A$ and $B, D_{a}$ and $D_{b}$, which are guaranteed to be bounded under threshold $D_{\text {th }}=400 \mathrm{~ms}$ with a varying network load. The TA-MAC reserves the minimum


Fig. 11. Aggregate network throughput under different network load conditions.
amount of time slots for both areas $A$ and $B$ to achieve the minimal average end-to-end delay for pairs of end users. Similarly, the analytical results match the simulation results.

Fig. 11 shows the aggregate network throughput for the TA-MAC versus the number of nodes. The throughput continuously increases with $N_{a}$, and the simulation results match well with the analytical results. A higher network throughput is achieved in the higher network load condition $\left(N_{c}=25, N_{b}=\right.$ 15 ), with more packets transmitted in each superframe.

## C. Performance Comparison

We compare the performance of the TA-MAC scheme with that of two existing MAC schemes proposed for multihop


Fig. 12. Average end-to-end packet delay comparison between the TA-MAC scheme and other MAC schemes.

MANETs: 1) a load adaptive MAC (LA-MAC) scheme [16] and 2) a DTSA [15], [16]. LA-MAC is a hybrid MAC scheme, in which each node is allocated one time slot for exclusive use based on a dynamic assignment. If current slot owner has no packets to transmit, all its two-hop neighbors can contend for the transmission opportunity in its designated time slot, based on a mechanism similar to the CSMA/CA. If a node fails transmission attempts for a consecutive number of times, referred to as the state switch threshold, the node switches to a high contention state and broadcasts a notification message, allowing its one-hop neighbors to contend in its designated time slot. This state switch is to reduce packet collisions caused by hidden nodes in a high traffic load condition. Additional simulation parameters for the LA-MAC are set as follows. Each backoff slot duration is $20 \mu \mathrm{~s}$. Both minimum and maximum contention window sizes (in the unit of backoff slot time) for slot owners and non slot owners are 1 and 8 backoff slots, and 9 and 16 backoff slots, respectively. The state switch threshold and packet retransmission limit are 5 and 7 , respectively, and the high contention state duration is 100 superframes. The DTSA is a dynamic TDMA scheme, where each node in a two-hop network is allocated one exclusive time slot within a time frame. The first slot in each frame is reserved for new nodes broadcasting request messages to join the network. If the current frame has no available time slot for newly arriving nodes, the whole frame length is doubled to generate new available time slots. Thus, the scheme guarantees each node occupying two time slots at most in every time frame.

Fig. 12 shows the comparison of average end-to-end delay versus the network load, with the same numbers of nodes in each area, between the TA-MAC scheme and the other two schemes. The LA-MAC achieves a smaller end-to-end delay in a low traffic load condition since nodes can contend to exploit the transmission opportunities in those time slots not used by the slot owners, making the MAC scheme behaving like CSMA/CA. However, the increase of node numbers in areas $A, C$, and $B$ results in a reduced number of empty slots and accumulated contention collisions for the LA-MAC, making it similar to the DTSA. Since the node time slot allocation in DTSA is not optimized, the traffic of relay nodes becomes quickly saturated, making the average end-to-end delay for both LA-MAC and DTSA increase dramatically to a large value in high network load conditions, whereas the TA-MAC


Fig. 13. Aggregate throughput comparison between the TA-MAC scheme and other MAC schemes.
achieves consistently minimal average end-to-end delay within a wide range of network traffic load. Its advantage becomes more obvious with a high number of nodes in the network by maintaining the end-to-end delay at a low level. The aggregate network throughput comparison is plotted in Fig. 13. The throughput of all three schemes consistently increases with the traffic load. In a low network load condition, all the schemes achieve similar channel utilization. However, when the network load increases, the proposed TA-MAC scheme achieves consistently higher throughput than the other two schemes, by optimizing the scheduling of token rotation cycles for each token ring and controlling the queue of each node in an unsaturated condition, whereas the throughputs for both LA-MAC and DTSA start to saturate from a moderate network load condition.

## VII. Conclusion

In this paper, we propose a distributed token-based adaptive MAC scheme for a two-hop IoT-enabled MANET. To eliminate the hidden terminal problem, a TDMA-based superframe structure is proposed to accommodate packet transmissions from different groups of nodes in separate TDMA durations. In each individual node group, a token is circulated probabilistically among the node members for distributed time slot allocation which is adaptive to variations of the numbers of nodes in each network area. To optimize the design parameters of the TA-MAC scheme, performance analytical models are developed in closed-form functions of the MAC parameters (i.e., the numbers of token rotation cycles scheduled for each token ring and the superframe length) and the network traffic load. Then, an optimization framework is proposed to minimize the average end-to-end delay by acquiring the set of optimal MAC parameters for each superframe. Analytical and simulation results demonstrate that the TA-MAC scheme achieves consistently minimal average delay for end-to-end transmissions, bounded delays for local transmissions and high aggregate throughput with variations of the number of nodes in the network. Based on a comparison with other two MAC schemes, the TA-MAC demonstrates much better scalability for the IoT-based two-hop environment in presence of network load dynamics, especially in a high traffic load condition.

## Appendix A <br> Proof of Proposition 1

For brevity, we only provide the proof for $D_{a}$. The proofs for other average delay functions can be carried out in a similar way. From (13), $D_{a}$ is the combination of average queueing delay, $D_{q}$, and average service delay $D_{t}$. That is,

$$
\begin{align*}
D_{a} & =D_{q}+D_{t} \\
& =\frac{\lambda_{a}}{2} \cdot \frac{x_{1} k_{a}^{2}+x_{2} k_{a}+x_{3}}{k_{a}+x_{4}}+\frac{\varepsilon_{a}}{k_{a}} \\
& =\frac{\lambda_{a}}{2} f_{1}\left(k_{a}\right)+f_{2}\left(k_{a}\right) \tag{21}
\end{align*}
$$

where $x_{1}, x_{2}, x_{3}$, and $x_{4}$ equal to the corresponding values in (13) $\left(x_{1}=\alpha_{a}, x_{2}=\beta_{a}, x_{3}=\gamma_{a}\right.$, and $\left.x_{4}=-\lambda_{a} \varepsilon_{a}\right)$.

In (21), $f_{2}\left(k_{a}\right)$ is a strictly convex function of $k_{a}$, since $f_{2}^{\prime \prime}\left(k_{a}\right)>0, \forall k_{a} \geq 1$. On the other hand, the second-order derivative of $f_{1}\left(k_{a}\right)$ can be derived as

$$
\begin{equation*}
f_{1}^{\prime \prime}\left(k_{a}\right)=\frac{2 x_{1} x_{4}^{2}-2 x_{2} x_{4}+2 x_{3}}{\left(k_{a}+x_{4}\right)^{3}}=\frac{g_{1}\left(x_{4}\right)}{\left(k_{a}+x_{4}\right)^{3}} \tag{22}
\end{equation*}
$$

Theoretically, $x_{1}, x_{2}$, and $x_{3}$ are fixed with a certain number of nodes, $L_{a}$, in area $A$, and $k_{a}$ can take values from the interval $\left[1,\left(M-L_{a c}-L_{b c} / L_{a}\right)\right]$, due to constraints (14a) and (14f). Thus, by conforming to constraint (14d), we have $x_{4} \in\left(-\left(M-L_{a c}-L_{b c} / L_{a}\right), 0\right)$ with the variation of $\lambda_{a}$, to guarantee $\left(k_{a}+x_{4}\right)^{3}>0$ in (22). The numerator of (22) can be regarded as a quadratic function of $x_{4}$, denoted by $g_{1}\left(x_{4}\right)$. We define $\widetilde{g_{1}\left(x_{4}\right)}$ as an extension of $g_{1}\left(x_{4}\right)$ with $\operatorname{dom} \tilde{g_{1}} \in(-\infty, \infty)$. Since $x_{1}>0$ and $x_{2}<0, g_{1}\left(x_{4}\right)$ represents a parabola, opening upward with the horizontal axis coordinate of its vertex $x_{4}^{*}=-\left(5 L_{a}^{2}+12 M L_{a}+1 / 12 L_{a}^{2}\right)$. Because $x_{4}^{*}<-\left(M-L_{a c}-L_{b c} / L_{a}\right)$, it is concluded that $g_{1}\left(x_{4}\right)$ with $\operatorname{dom} g_{1} \in\left(-\left(M-L_{a c}-L_{b c} / L_{a}\right), 0\right)$ is a strictly increasing function of $x_{4}$. Furthermore, since $g_{1}\left(-\left(M^{-}-L_{a c}-L_{b c} / L_{a}\right)\right)>$ 0 , it is proved that $g_{1}\left(x_{4}\right)>0, \forall x_{4} \in \operatorname{dom} g_{1}$, and thus we have $f_{1}^{\prime \prime}\left(k_{a}\right)>0, \forall k_{a} \in\left[1,\left(M-L_{a c}-L_{b c} / L_{a}\right)\right]$. Hence, $D_{a}$ is a linear combination of two strictly convex functions of $k_{a}$, which is also known to be strictly convex [30].

## Appendix B

Proof of Corollary 1
According to Proposition $1, D_{a}$ is a convex function of $k_{a}$. Thus, we have $D_{a}^{\prime \prime}\left(k_{a}\right)>0$, indicating that $D_{a}^{\prime}\left(k_{a}\right)$ is a strictly increasing function which is derived as

$$
\begin{align*}
D_{a}^{\prime}\left(k_{a}\right) & =\frac{\lambda_{a}}{2} \cdot \frac{x_{1} k_{a}^{2}+2 x_{1} x_{4} k_{a}+x_{2} x_{4}-x_{3}}{\left(k_{a}+x_{4}\right)^{2}}-\frac{\varepsilon_{a}}{k_{a}^{2}} \\
& =\frac{\lambda_{a}}{2} f_{3}\left(k_{a}\right)+f_{4}\left(k_{a}\right), \quad k_{a} \in\left[1, \frac{M-L_{a c}-L_{b c}}{L_{a}}\right] . \tag{23}
\end{align*}
$$

Therefore, the maximum value of $D_{a}^{\prime}\left(k_{a}\right)$ is obtained when $k_{a}=\left(M-L_{a c}-L_{b c} / L_{a}\right)$. That is,

$$
\begin{align*}
D_{a}^{\prime}\left(\frac{M-L_{a c}-L_{b c}}{L_{a}}\right)= & \frac{\lambda_{a}}{2} f_{3}\left(\frac{M-L_{a c}-L_{b c}}{L_{a}}\right) \\
& +f_{4}\left(\frac{M-L_{a c}-L_{b c}}{L_{a}}\right) \tag{24}
\end{align*}
$$

In (24), $f_{3}\left(\left(M-L_{a c}-L_{b c} / L_{a}\right)\right)$ is a function of $x_{4}$, with $x_{4} \in$ $\left(-\left(M-L_{a c}-L_{b c} / L_{a}\right), 0\right)$, which is expressed as

$$
\begin{align*}
f_{3}( & \left.\frac{M-L_{a c}-L_{b c}}{L_{a}}\right) \\
& =\frac{\left[\frac{2 x_{1}\left(M-L_{a c}-L_{b c}\right)}{L_{a}}+x_{2}\right] x_{4}+x_{1}\left(\frac{M-L_{a c}-L_{b c}}{L_{a}}\right)^{2}-x_{3}}{\left(\frac{M-L_{a c}-L_{b c}}{L_{a}}+x_{4}\right)^{2}} \\
& =\frac{g_{2}\left(x_{4}\right)}{\left(\frac{M-L_{a c}-L_{b c}}{L_{a}}+x_{4}\right)^{2}} . \tag{25}
\end{align*}
$$

Since $\left(2 x_{1}\left(M-L_{a c}-L_{b c}\right) / L_{a}\right)+x_{2}<0$, the linear function $g_{2}\left(x_{4}\right)$ is a strictly decreasing function with its maximum value being $g_{2}\left(-\left(M^{-}-L_{a c}-L_{b c} / L_{a}\right)\right)<0$. Thus, we have $f_{3}\left(\left(M-L_{a c}-L_{b c} / L_{a}\right)\right)<0$, and $D_{a}^{\prime}\left(M-L_{a c}-L_{b c} / L_{a}\right)<0$, $\forall x_{4} \in\left(-\left(M-L_{a c}-L_{b c} / L_{a}\right), 0\right)$. Hence, it is proved that $D_{a}^{\prime}\left(k_{a}\right)<0, \forall k_{a} \in\left[1,\left(M-L_{a c}-L_{b c} / L_{a}\right)\right]$. Similar proofs for the same property of other one-hop average delay functions can be made, which are omitted here.

## Appendix C <br> Proof of Proposition 2

If we rewrite (SP2a) in a standard form, it is easy to verify that all the inequality constraint functions are convex with respect to $k_{a c}$. In the objective function, $D_{a b}(\cdot)$ is a summation of $D_{a c}(\cdot)$ and $D_{c b}(\cdot)$ according to (11), which is further expressed as

$$
\begin{equation*}
D_{a b}\left(k_{a c}, h\left(k_{a c}\right)\right)=D_{a c}\left(k_{a c}\right)+D_{c b}\left(h\left(k_{a c}\right)\right) \tag{26}
\end{equation*}
$$

Based on Proposition 1, we know that $D_{a c}(\cdot)$ is a convex function of $k_{a c}$ and $D_{c b}(\cdot)$ is convex function of $h\left(k_{a c}\right)$. It is also found that $h\left(k_{a c}\right)$ is a linear function of $k_{a c}$. Thus, according to the scalar composition rules, $D_{c b}\left(h\left(k_{a c}\right)\right)$ is also a convex function of $k_{a c}$. Hence, $D_{a b}(\cdot)$, a linear combination of two convex functions, is also convex with respect to $k_{a c}$. The same property also holds for $D_{b a}(\cdot)$ with a similar proof. Moreover, as stated before, the 2-D max function, $\max \{x, y\}$, is convex on $\mathbf{R}^{2}$ and nondecreasing in each of its two arguments. Therefore, according to the vector composition rules, the objective function of (SP2a) is a convex function with respect to the decision variable $k_{a c}$, which ends the proof.

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[^1]:    ${ }^{1}$ An end user can also mean an end device in this paper

[^2]:    ${ }^{2}$ To smooth the delay jitter, we assume that each backlogged node transmits one data packet each time that the node holds the token.

[^3]:    ${ }^{3}$ Because of the geographical symmetry of areas $A$ and $B$, we only consider $S(D)$ nodes moving between areas $A$ and $C$. Similar results can be obtained when nodes move between areas $B$ and $C$.
    ${ }^{4}$ Since nodes move at a low speed, we assume that a token holder can pass the token to the next token holder before moving to a new network area.

[^4]:    ${ }^{5}$ Since traffic arrival rates at each relay node are greater than that at each source node, queue stability of source nodes in $R_{a c}$ and $R_{b c}$ is also guaranteed by constraint (14c).

