Improving Packet Delivery Fraction of Ad Hoc Network with XORs-based Network Coding CDS

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Abstract—This paper addresses the subject of improving performance of a client / server ad hoc network system suffering from high resource contention by improving its packet delivery fraction. In previous work authors gave a solution to reduce the possibility of bottleneck occurrences for such a system, which main idea is to decouple route-requesting from serviceproviding activities, and try to reduce the former. This was achieved by that for each busy, and thus critical server, a node cluster is created by constructing a server-centered Connected Dominating Set (CDS) to actively reduce the contention level. This paper presents an improved solution utilizing the network coding technique with constructing the server-centered XORsbased network coding CDS. It is found that in this way, the overall data packet delivery fraction can be improved greatly furthermore. Results of simulation experiments demonstrated the effectiveness of the proposed approach.

Index Terms—ad hoc network, server-centered CDS, network coding, resource contention, packet delivery fraction.

I. INTRODUCTION

N ad hoc network is an autonomous collection of mobile nodes that communicate over wireless links. Each node functions as both a terminal and a router, i.e., the routing functionality is its constituent element. Each node is free to move in any direction at any time, so the network's topology may change unpredictably over time.

For many applications on an ad hoc network, nodes can be divided into two classes according to their roles playing in applications: the server that provides services and the client that requests services provided by servers. For this, an example is given below.

Example 1 (client-server environment on ad hoc network). Assume a crowded ad hoc network with a lot of nodes, where a teacher and a class of students, each one being with a mobile computer, are taking an outdoor seminar. The teacher's computer is used to coordinate the seminar, and to take the responsibility for storing data used, questions raised, and reports submitted by students. Obviously that computer would play the role of a server, and others would be clients.

It was found that for such a system, it is the server and its adjacent nodes that dominate performance, and tend to cause bottlenecks due to congestion. To reduce the possibility of bottleneck occurrences, we suggest relieving resource contention of the area where a server node is located. In fact, no matter it plays a role of server or client, each node has to assist other nodes with discovering routes. Then in a service intensive environment such as that given in Example 1, this extra task enforcing on the server node would degrade system performance [1]. Especially, if reactive (on-demand) routing





Fig. 1. An example for showing how the bottleneck can occur in an ad hoc network. In this example, the Node 8 is functioning as a server, and others are clients.

protocols such as AODV (Ad hoc On-Demand Distance Vector) [2] is used, the performance down would become more serious.

To further explain the problem, let us consider the network shown in Fig. 1. Assume that node 2 is to exchange data with node 17. If there is not any pre-hand knowledge about the route from node 2 to node 17, all nodes around node 8, the server node, has to actively take part in the process of discovering routes. This may burden on server node too much. This also causes another serious problem: more collision coming from interference. It was said that due to interference, end-to-end throughput would be as low as only 4.2% the link data rate in the worst case [3]. As results, performance of the server, and therefore the system, would be affected seriously.

Then in order that a system can give satisfactory performance even in the case of high resource contention, on the one hand, we have to know which nodes are playing the role of servers. For this, an approach to "mining" the server nodes on demand has been proposed in [4]. On the other hand, when server nodes are located, we have to ease the resource contention around them. For this, possible solutions include topology control based on transmission power reduction [5]– [7] and dominating-set-based routing [8]–[13]; A CDS-based approach has been proposed in [4]. This paper continues the work in [4] along the line of CDS-based solutions.

Dominating-set-based routing is such a solution that selects certain nodes from the network as gateway nodes. These gateway nodes form a Connected Dominating Set (CDS) and are responsible for routing within network. In [4] authors introduced the idea to construct CDSes centered around server nodes, so that other nodes of a CDS can take the place of the server node to conduct routing, to prevent a server node from excessive routing activities. In this paper, authors extend the idea, and present a revised version that takes advantage of the network coding technique.

Network coding is such a technique where, instead of simply relaying packets they receive, nodes of a network will combine several packets together for transmission so as to improve the capacity of multi-hop wireless networks [14]–[17]. "The potential advantages of network coding over routing include resource (e.g., bandwidth and power) efficiency, computational efficiency, and robustness to network dynamics" [18]. Considering its ability to improve network capacity, we think network coding should be able to help to ease the resource contention.

As stated above, this paper departs from our previous work in [4] in that an extended version based on the network coding technique is proposed. In this paper we address the subject of improving performance of an ad hoc network system suffering from high resource contention by improving its packet delivery fraction when necessary. It is suggested that resource contention of the area where a server node is located be relieved. This is achieved mainly by decoupling route-requesting from service-providing activities, and reducing the former around the servers. On the one hand, for each busy, and thus critical server, a node cluster is created by constructing a server-centered network-coding CDS to actively reduce the level of resource contention. On the other hand, routes are adaptively selected to avoid high congestion area. It is found that in this way, the overall packet delivery fraction can be improved greatly. Results of simulation experiments demonstrate the effectiveness of the proposed approach.

The remainder of this paper is structured as follows. In Section II, we describe the model of computation and formally define the problem. An algorithm for solving the problem defined in Section II is given in Section III. Results of a performance study is discussed in Section IV. Finally, Section V concludes the paper and indicates directions for future research.

II. PROBLEM STATEMENT

The problem to be solved in this research, as described in Section I, is formulated as follows.

Problem Statement. Given an ad hoc network of mobile nodes, in which each node can be categorized as either server or client. An approach is to be devised to improve the overall system performance by relieving working load of the server nodes when necessary. Taking Fig. 2 as an example, this means that when number of source nodes reaches about 30, some means are taken to avoid the sudden falling of packet delivery fraction shown in this figure.

One possible solution to this problem is to construct a "global" CDS that covers the whole ad hoc network. This dominating-set-based approach is widely used for constructing network backbones. However, it could be very expensive to maintain such a CDS dynamically to adapt topology changes due to the mobility of mobile nodes, which is a main property of ad hoc networks. It is true that various low cost approaches have been proposed [12], [19]–[26]. For the defined problem, however, these approaches could not be very effective. It is the lightweight and diameter-limited CDS that



Fig. 2. Simulation results showing that packet delivery fraction may fall suddenly due to resource contention.

is required here. On the one hand, as stated above, existing dominating-set-based approach requires exchanging a lot of messages to construct and maintain a CDS, which contradicts our objective: ease the resource contention. On the other hand, considering our objective, to limit the diameter of a CDS to some specified threshold, as proposed in [27]–[30], is necessary. "With the help of a CDS with small size and diameter, routing is easier and can adapt quickly to topology changes of a network" [27]. Differently, we require that a CDS should be centered at a server node, and its structure should be suitable for the defined problem. That is, we focus our attention on the construction of a service-providing CDS. More attention should be paid to the internal structure of the CDS. An effective approach has been proposed in [4].

Another possible solution to this problem is to take advantage of the network coding technique. Li et al. [31], [32] investigate the benefit of network coding over routing for multiple independent unicast transmissions. Ho et al. [33] consider network coding across multiple unicasts within the class of network codes restricted to XOR coding between pairs of flows. Chachulski et al. develop MORE [34], an opportunistic routing protocol that exploits intra-flow network coding. An opportunistic XOR coding is proposed [35], [36], which is an inter-flow network coding protocol, and packet mixing is performed at the joint nodes of the paths determined by the routing module. Methods proposed in [37], [38] seem to be able to provide more opportunities to mix traffic flows. We distinguish our work in this paper in that we incorporate network coding into the server-centered CDS for independent transmission sessions, and focus on relieving resource contention to improve the packet delivery fraction.

III. RELIEVING LOAD OF THE CRITICAL NODES

In this section, we first provide notations, definitions, and some fundamental concepts used in the subsequent sections. Then we propose an approach to identifying server nodes and relieving their loads. Figure 3 shows the system model, and Figure 4 shows the resulting CDS corresponding to Fig. 3. We will use these two figures to give our description.



Fig. 3. System structure. Node 0 is a server node, and others are clients.



Fig. 4. Server-centered CDS would be a tree that takes a server node as its root. Numbers in circles are node ids. Numbers to the left are level numbers.

A. Definitions and Preliminaries

Let graph G = (V, E) represent an ad hoc network, where V consists of all nodes and E represents all the links. Without losing generality, the nodes in V are assumed to be located in a plane, and each node $v_i \in V$ has a transmission range r. An edge $(v_i, v_j) \in E$, iff $d(v_i, v_j) < r$, where $d(v_i, v_j)$ denotes the distance between v_i and v_j .

Firstly we define the term *server node*. For any node $v \in V$, the number of inbound route requests to v for time period $T = [t_0, t_1]$ is denoted as IN(v, T), and the number of outbound route requests originating from v is denoted as OUT(v, T). The number of inbound route requests, including forwarded ones, made by nodes within 3 hops from v (i.e., locally) is denoted as $IN_{\leq 3}(v, T)$, and that made by nodes 4 hops away (i.e., globally) is denoted as $IN_{>3}(v, T)$. The number of lost packets due to collision and congestion is denoted as $Coll_Cong(v, T)$. Then server node is determined according to the following definition.

Definition 1 (Server node). A node $v \in V$ is taken as a server node for time period of T iff it satisfies the following three conditions, where λ_1 , λ_2 , and λ_3 are threshold values defined by users beforehand.

1) v satisfies the following inequality, and thus is consid-

ered as being functioning as a service provider,

$$\frac{IN(v,T)}{IN(v,T) + OUT(v,T)} \geq \lambda_1$$

2) v satisfies the following inequality, and thus is considered as being involved in processing route requests coming from global nodes.

$$\frac{IN_{>3}(v,T)}{IN_{\leq 3}(v,T) + IN_{>3}(v,T)} \geq \lambda_2$$

3) *v* satisfies the following inequality, and thus is considered being suffering from resource contention.

$$PPLost = \frac{Coll_Cong(v,T)}{TX_Rcv(v,T) + Coll_Cong(v,T)} \ge \lambda_3$$

Figure 3 shows a part of an ad hoc network. Node 0 is a server node. Nodes 23-33 are taken as *global nodes* and others are taken as *local nodes*.

Next we define the term *server-centered CDS*. A set of nodes $V' \subseteq V$ forms a Connected-Dominating Set (CDS) of network G, if all members in V - V' are neighbors of a node in V' and the subgraph G[V'] induced from V is connected. A server-centered CDS, denoted as *SCDS*, is a CDS that satisfies the following definition.

Definition 2 (Server-centered CDS: SCDS(v)). *Given a* server node v, the v-centered CDS, denoted as SCDS(v), is a CDS of diameter 3. It takes v as its kernel for providing services, layer 3 as its interface for accepting requests, and layers 1 and 2 as the buffering layer for resisting interferences from interface layer and outer nodes, and forwarding packets.

In Fig. 3, nodes represented by bold black circles form a server-centered CDS, the *SCDS(0)*.

We adopt the definition of d-CDS in [39] as follows. G_d denotes the *d*-closure of G. It has the same node set as G, but the edge set is: $\forall u, v \in V$, there is an edge connecting u and v iff $0 < \delta(u, v) \leq d$, where $\delta(u, v)$ is the distance between u and v. A set $V' \subseteq V$ is a *d*-hop dominating set of G if it is a dominating set for G_d . That is, if every node of Gis within a distance d of some node in V'. A 1-dominating set is simply called a dominating set. We say that V' is *d*-hop connected if it is connected in G_d .

Then according to the above definition of d-CDS, and referring to Fig. 2 and Fig. 3, it is obvious that the following property holds true for a server-centered CDS. More description will be given in Section III-B.

Property 1. A server-centered CDS is a 1-CDS.

In remainder of this section we define the server-centered network coding CDS. A part of Fig. 4 is redrawn in Fig. 5. As shown in this figure, across node 1 we would have multiple transmissions. Among them, some are between node 0 and node 1 (identified by $S_{0,1}$), and others are between node 1 and node 4 (identified by $S_{1,4}$), or node 1 and node 5 (identified by $S_{1,5}$). For these vertical transmissions, more deterministic coding is possible. There are also horizontal transmissions such as that identified by $S_{1,2}$, for which we may also benefit from the opportunistic coding. So we have the mixed coding scheme.

A XOR-based coding similar to that in [36] is adopted. For one example, at node 1, when a packet from node 0 is



Fig. 5. Server-centered Network Coding CDS. $S_{i,j}$ means the transmission between node i and node j.

received and its next hop is node 4, and in output queue a packet from node 4 is found and its destination is node 0, then node 1 XORs the two packets and broadcasts the XORed one. Note that we are considering the server-centered CDS, and there would have a lot of such transmissions. For another example, a downward packet sent to node 1 for relay is overheard by node 2, and an upward packet received at node 1 is needed to forward to node 2, then node 1 XORs the two packets and broadcasts the XOR-ed one.

Definition 3 (Server-centered network Coding CDS: SC-CDS(v)). Given a server node v, a v-centered network coding CDS, denoted as SCCDS(v), is a SCDS that make use of network coding from two directions, the vertical coding for deterministic coding, and the horizontal coding for opportunistic coding.

B. The Algorithm SCCDSR: SCCDS-based Routing

This section gives an approach to solving the above defined problem, including functions for collecting system status data (Func. 1), constructing server-centered CDS (Func. 2), forwarding packets (Func. 3), coding data packets, and avoiding busy routes (Func. 4). About details of Func. 1, 2, 3, and 4, see also [4]. Table 1 summarizes the notations that will be used.

1) Determining the server node: For every node in an ad hoc network, data concerning system status defined in Definition 1 will be collected periodically according to the algorithm shown in Func. 1 below. These data will be used to decide which node is a server node, and when to start constructing server-centered CDS. The parameters $\lambda_1 - \lambda_3$

TABLE I NOTATIONS USED IN THIS PAPER

notations	meanings
x.id	identity number of node x
x.level	level number of node x
x.father	identity number of node x's father
v.adj	adjacent nodes of node x
x.sons	sons of node x
$path_to(s)$	route to server node s
$Get_F(v)$	return id of v's father node
sender(p)	return id of the node that sends packet p
F(v)	fathers of node v
broadcast(m, f, p)	broadcast message m,
	id f of the current node's father,
	and path p to the server node

Function 1 Get system status
Require: Time interval T
Ensure: Each node knows system status within 3 hops
1: for (each node $v \in V$) loop
2: for (each time interval T) do
for (each sent, received, and lost packet) update
\cdot IN, OUT,
$\cdot IN_{\leq 3}, IN_{>3},$
· TX_Rcv, Coll_Cong
3: end for
4: end for
5: end for

in Definition 1 can be determined beforehand by considering characteristics of the application environment. For example, one can set λ_1 , λ_2 , and λ_3 to 0.9, 0.5, 0.3, respectively.

2) Constructing server-centered network coding CDS: When a node becomes a server node according to Definition 1, it is the time to construct the server-centered network coding CDS. An algorithm is given in Func. 2.

When a SCCDS is constructed, each node of it will know the path to the server node s. For example, in Fig. 4, node 13 knows the route to node 5, 1, and 0.

Function 2 Construct SCCDS
Require: A server node s identified by s_{id}
Ensure: A <i>s</i> -centered CDS, <i>SCCDS</i> (<i>s</i>)
1: for (each $v \in V$) do $v.level \longleftarrow \infty$;
2: $s.level \leftarrow 0;$
3: $SCCDS(s) \leftarrow \{ \langle s.id, s.level \rangle \};$
4: $m_0 \leftarrow \langle s.id, s.level \rangle;$
5: broadcast(m ₀ , null, null);
6: for (each $v \in V$ receiving m_0 and $v.level = \infty$) do
7: $v.father \leftarrow s.id;$
8: $v.level \leftarrow 1;$
9: $m_1 \leftarrow \langle v.id, v.level \rangle;$
10: $broadcast(m_1, v.father, path_to(s));$
11: end for
12: $l \leftarrow 1;$
13: repeat
14: for (each $v \in V$ receiving m_l) do
15: switch ($v.level$) {
16: case $l - 1$:
17: if $(get_F(sender(m_l)) = v)$
18: then $v.sons \leftarrow v.sons \cup \{ sender(m_l) \};$
19: case l :
20: if $(l > 3)$ then break;
21: $v.adj \leftarrow v.adj \cup \{ sender(m_l) \};$
22: case ∞ :
23: if $(l > 3)$ then break;
24: $v.level \leftarrow l+1;$
25: $m_{l+1} \leftarrow \langle v.id, v.level \rangle;$
26: $father \leftarrow min_id(F(v));$
27: $broadcast(m_{l+1}, father, path_to(s));$
28: endswitch
29: end for
30: $l \leftarrow l+1;$
31: until ($l > 4$);
32: $SCCDS(s) \leftarrow SCCDS(s) \cup \{ v \}_{v.sons \neq null};$

Function 3 SCCDS-based routing

Require: SCCDS(s) and an inbound route request packet p **Ensure:** The request is processed within SCCDS(s)1: if (n is for server node s.) then

1:	If (p is for server node s) then
2:	return the route to s directly;
3:	else
4:	repeat
5:	try (forward p to) {
6:	· a level-3 node;
7:	· a level-2 node if no adjacent level-3 nodes;
8:	}
9:	if (no level-2 and -3 nodes are available)
10:	then break;
11:	else if (find a route, say r) return(r);
12:	until (p has traveled k level-1 subtrees);
13:	broadcast p and find route within the remaining level-1
	subtrees using the general method as AODV:

14: end if

Note that two nodes apart from no more than 3 hops may start constructing SCCDS at the same time. At that time, we may simply stop constructing process of the node with a smaller node id to resolve the confliction. We may also compare the metrics defined in Definition 1 to determine to stop which one more smartly.

A SCCDS constructed with using Func. 2 will be a 1-CDS. This is because some interior nodes of the tree constructed in Func. 2 are taken as the CDS nodes, and all leaf nodes are removed.

3) SCCDS-based routing: Without losing generality, we assume that the method of discovering a route is the same as that of AODV, except a route request is in the SCCDS, in which case it is processed according to Func. 3.

The idea of Func. 3 is that route requests that are not for nodes within the CDS are forwarded along the outer layer as far as possible so as to avoid interference to the server node. However, if there is no such nodes available, a shortest path across the server node is taken so that the packet can get out of the CDS as soon as possible.

The optimal value of parameter k in Func. 3 is determined to be 2 by experiments. We had expected that a larger kwould give better performance, but by experiments we found that a k larger than 2 would cost too much.

4) Coding Data Packets: A data packet received for forwarding to other nodes within a SCCDS will be coded if possible, as defined in Definition 3. For packets concerning route requests, coding will not be applied. This is because it was found that coding route requests would not improve, in some cases would degrade packet delivery fraction.

The coding algorithm follows from Definition 3. As [36] we use the easy-to-implement and relatively inexpensive XOR operation to perform coding. At each CDS node, mainly two packet-storing facilities are involved: the output queue for keeping the packets to be sent out or forwarded, and the packet pool for buffering the packets overheard on the CDS or sent out in the past T_{buf} seconds.

As [36] we also adopt the principle of *never delaying packets*. But instead of checking encoding opportunities when the wireless channel is available, we do that when a packet is to be inserted into the queue. On the one

hand, we found that a low packet delivery fraction is often resulting from the queue overflow due to too many packets are received [4], and our design decision helps to reduce the queue length. On the other hand, our objective is to relieve resource contention, and a short or null queue means low resource contention, implying that the coding is not necessary so that the corresponding cost can be avoided.

5) Avoiding the busy routes: While a node is exchanging data with some other node using a known route, a new route may become available to it. In that case, we have to make decision which one should be used to get better performance such as higher throughput, higher delivery fraction, and so forth. This decision is made according to Func. 4. By experiments it was found that the optimal value of parameter α is between 0.7 to 0.8, and was set to 0.75 in simulation experiments.

Function 4 Process a route response

Require: A packet responding to a route request **Ensure:** A more suitable route is elected.

 When a packet responding to a route request goes back to the node that has made the request, let the route be <v₀, v₁, ..., v_n>, it collects data about *PPLost_i*, and computes the survival probability of a data packet that will be sent along the route as follows:

$$PPSurv = \prod_{i=1}^{n} (1 - PPLost_i)$$

2: if (*PPSurv* of the current route $\leq \alpha$ and

PPSurv of the new route > that of the current route) **then** replace the route with the new one;

IV. PERFORMANCE EVALUATION

Simulation experiments have been done to study performance of the proposed approach. We take the AODV algorithm, which is known to be of very high packet delivery fraction, as a baseline, and evaluate to what extent the proposed approach can improve system performance in the condition of high resource contention. This section describes simulation environment, presents and discusses simulation results.

A. Simulation Environment

Simulation experiments are done using Network Simulator-2. As shown in Fig. 6, the field configuration is 2000 m \times 1500 m square with total 100 nodes. Nodes are randomly placed on the plane. Continuous Bit Rate (CBR) traffic sources are used for traffic model. Data packet size is 512-byte. The number of source-destination pairs is varied to change the offered load. The number of source nodes is 10, 20, 30, 40, and 50 separately. The MAC we employ is 802.11 MAC.

Three performance metrics, as described below, are used for the performance study:

- Packet Delivery Fraction: The ratio of total data packets successfully received to total ones sent by CBR sources.
- Average Path Length: The average length of all paths that connect source and destination nodes.

• Number of Router Drops: Total number of data packets dropped by the routers. This routing load metric demonstrates the level of resource contention and also evaluates the efficiency of the routing protocol.

In order to more clearly demonstrate the effectiveness of the proposed approach, we disable the mobility of nodes in this performance study. For the same reason, traffic model is so designated that the node 0 in Fig. 6, located at the centerleft of the figure, could be always selected as the server node.

B. Experimental Results and Discussions

Figure 7 gives results of packet delivery fraction vs. number of source nodes. Line AODV-30 (resp. AODV-50) is for the original AODV in the case that 30% (resp. 50%) accesses from source nodes are for the server node. Lines SCDSR-30 and SCDSR-50 are for the proposed approach. Line NC-SCDSR-50 is for SCDSR-50 with the network coding, i.e., the SCCDSR proposed in this paper.

From Fig. 7 we know that on the one hand, SCDSR can effectively improve packet delivery fraction in the case of high resource contention. In the case of 30% (resp. 50%) accesses going to a server node, it begins to outperform the original AODV when number of source nodes becomes larger than 35 (resp. 31), and raises packet delivery fraction from about 64% (resp. 60%) to about 75% (resp. 80%) when number of source nodes is 50. On the other hand, the proposed network coding-based approach in this paper, the SCCDSR, does have benefits, and can substantially improve packet delivery fraction even more than SCDSR does. From the same figure we also know that if the level of resource contention is low, the original AODV would give higher performance with respect to the packet delivery fraction.

We think that the main reason of the results shown in Fig. 7 lies in the proposed approaches' ability to reduce resource contention, which is demonstrated in Fig. 8 and 9. Figure 8 shows that, numbers of packets dropped at router nodes are lower for the proposed approach, implying that the proposed approach really has the ability to reduce resource contention. As shown in Fig. 9, when the level of resource contention becomes high, the average lengths of paths become long due to the adaptive route redirection mechanism of AODV to avoid the congestion-intensive routes. This increased average path in turn increases the packet delivery fraction. Compared with the original AODV, the proposed approach tends to



Fig. 6. The ad hoc network used in performance study.



Fig. 7. Results of packer delivery fraction vs. number of source nodes



Fig. 8. Results of number of router drops vs. number of source nodes



Fig. 9. Results of average length of paths vs. number of source nodes

reduces the resource contention, and thus the average length of paths.

For the reason of SCCDSR outperforming SCDSR in Fig. 7, we think this is mainly because network coding makes

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node's queue shorter, as described in Section III, reducing the probability that a node will drop packets that have already consumed network resources.

Figure 10 shows results of packet delivery fraction vs. number of source nodes in the case that each client node sends data to and receives data from the server node only. Figure 11 shows the results of average length of paths. The lines With-NC and Without-NC show results of SCCDSR and SCDSR respectively.



Fig. 10. Results of packer delivery fraction vs. number of client nodes



Fig. 11. Results of average length of paths vs. number of client nodes

For SCCDSR, these figures more clearly shows that packer delivery fraction improvement varies depending on the amount of traffic between client and server. Note that in this experiment, the traffic between one pair of client and server may be asynchronous; otherwise more improvement may be possible.

V. CONCLUSIONS

We have identified a performance problem in an ad hoc client / server system, and given a solution of it. The suggested system model divides nodes into two classes: servers and clients. For each server, a server-centered CDS (SCDS) is constructed. Each SCDS has a kernel, an interface layer, and a buffering layer. The CDS with such a structure, combined with the supporting algorithms, is found to be able to relieve load of the server node and avoid interference from outer nodes, and thus improve overall system performance. Simulation experiments have been conducted and experiment results demonstrated effectiveness of the proposed approach.

An unsolved problem is that sometimes a SCDS cannot accurately represent the physical distance. So interference may still reach server node. Solving this problem is our future work.

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