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Multi-hop D2D Communications with Network Coding: From a Performance Perspective

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Abstract—Multi-hop D2D communications play an important role in expanding D2D coverage. In this paper, we study a relay-based and network-coding-assisted (in particular, XOR coding) multi-hop D2D communication system. In the system, toward jointly considering the impact of interference and network traffic conditions on the quality of D2D communications, various channel fading models and traffic models are investigated, and the packet loss probability of D2D links is meticulously computed using these models. With packet loss probability of D2D links, the general closed-form expressions of end-to-end packet loss probability (E2EPLP) of the system with the presence (or absence) of XOR coding are subsequently derived. Our experiments reveal that both the number of relays and the mechanism of XOR coding can affect the system performance. Specifically, the increase in the number of relays will lower the overall system performance (e.g., an increase in the E2EPLP and end-to-end completion time, and a decrease of the end-to-end rate may follow as a result). On the other hand, although the presence of XOR coding unfortunately raises the system E2EPLP, it can effectively improve the end-to-end completion time and end-to-end rate. It is our belief that the analytical approach proposed in this paper and the results found in our work can be useful to peer studies in the context of applying network coding in multi-hop D2D networks.

Index Terms—multi-hop D2D communication, network coding, traffic model, performance evaluation.

I. INTRODUCTION

The paradigm of device-to-device (D2D) communications underlaying cellular networks has been proposed to address the dramatically increasing Internet access demands from mobile users. This paradigm is widely considered as an effective way to improve transmission rate, to reduce transmission latency and power consumption, and to enhance spectrum efficiency [1]. Unfortunately, the direct single-hop D2D communication mechanism only works well within a region of a short radius, and has trouble in delivering quality transmissions over a large area. Toward remedying this, the idea of multi-hop D2D communications has been proposed and investigated to expand the D2D coverage [2].

Relay nodes are used in multi-hop D2D communications to forward data packets. On the one hand, this mechanism can expand the signal communication range, which makes D2D communications be able to fit complex and diversified environments. On the other hand, it can also shorten the distance of D2D links per hop, which may eventually reduce the transmit power and energy consumption. Incidentally, the relay selection strategy is of paramount significance in multi-hop D2D communications. In the area of cooperative communications, relay techniques have been studied comprehensively. For example, [3] studied a social relationship to effectively select relay users, [4] proposed a social-aware energy-efficient relay selection mechanism that considers the hidden social ties among mobile users. Also, a relay selection policy based on the buffering ability of relay nodes was presented in [5], and the random distribution of relay nodes by treating the relay deployment as a Poisson point process was investigated in [6]. In this paper, we assume that relay users are able to and willing to forward data packets, and devise a primitive relay selection strategy that is based on the location information of users in the multi-hop D2D communication scenario. In this strategy, the base station selects the relay users in such a way that the sum of the distances from the selected relay to the two D2D ends is minimal.

It is known that network coding is capable of improving network throughput performance by allowing relay nodes to compute and then forward the received packets [7], [8], [9]. Generally speaking, the techniques of network coding can be divided into the following major categories [10]: physical layer network coding (PNC), analog network coding (ANC), instant decodable network coding (IDNC), and random linear network coding (RLNC). Each network coding method has its own pros and cons that fit best with a particular scenario.

Many efforts have been made to analyze the performance of network coding or relay assisted D2D communications. [11] analyzed the average energy efficiency and spectral efficiency by assuming that the signal fading is Rayleigh fading. [12] and [13] studied the problem of resource allocation for network coding aided D2D communications. The completion time required for recovering all missing packets at devices using IDNC was investigated in [14] and [15]. The authors of [16] developed a radio resource management (RRM) mechanism and investigated how caching capabilities at relay nodes would impact the system performance. While [17] evaluated the performance of the network coding aided D2D communications in terms of throughput, energy efficiency, and battery consumption, [18] devised a game-theoretic model for the compensation power acquisition of D2D transmitters. [19] considered a new class of codes (fountain codes) to keep reliable transmissions of data in multi-hop erasure channels. [20] proposed a reliable multicast protocol which is built upon opportunistic routing and random linear network coding. Furthermore, the

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closed-form expressions of outage probabilities were obtained for relay-assisted OFDM D2D systems in [21]. The network throughput and end-to-end packet loss probability (E2EPLP) were evaluated for multi-hop D2D communications in [22]. Differing from previous studies, we procure the generic closed-form expressions of E2EPLP with the presence or absence of the network coding, respectively, on the basis of the packet loss probability (PLP) of D2D links. Also, considering that some existing studies [22], [23] only investigate the special case of one relay node (two-hop) D2D communications, we extend the prior studies by considering the general case of an arbitrary number of hops in our work.

Note that network transmissions typically experience intermittent interference or noise causing unrecoverable errors in packets that are then discarded at the receivers, we need to choose a suitable channel fading model to analyze the PLP of D2D links. In most of the prior studies, the Rayleigh fading model, which ignores D2D communications in the line-of-sight (LoS), is adopted. This, in fact, is problematic in the sense that close-range communications (e.g., D2D communications) often lead to the existence of dominant LoS signal components in the received desired signals [24], [25], [26]. In addition, as packet losses occur frequently due to network traffic congestions that cause packet buffer overflows in relay nodes, there is a need to select an appropriate traffic model to analyze the queue capacity and the service capability of relay nodes in multi-hop D2D communications. This has also been extensively explored in the literature. For example, the packet arrival rate, when the packet arrival in D2D communications is treated as a Poisson arrival process, is studied in [27]; a dynamic D2D traffic model and the ensuing queue delay of D2D communications are investigated in [28]; and a dynamic data arrival process with non-saturated buffers is studied in [29]. As such, we in our work investigate the system performance in different channel fading models (Rician-Rayleigh and Rayleigh-Rayleigh) and in different traffic models (Pareto, fractional Brownian motion, and Poisson).

Our contributions in this paper are summarized as follows.

- A general multi-hop D2D communication model underlaying the cellular network is introduced, which is followed by the formulation of the related channel fading models and traffic models. We propose a primitive relay selection strategy, in which the base station selects the relay nodes using their polar coordinates information and by the criterion of the “shortest path” (see Section II-D). Moreover, a coding scheme where packets are encoded by network coding at the relay nodes is investigated for the data transmission over multi-hop D2D links.

- By jointly considering the impact of interference and network traffic conditions on the quality of D2D communications, the packet loss probability of D2D links is carefully constructed from two perspectives – the outage probability and packet failure probability, which are subsequently computed by using the Rician-Rayleigh fading model and by using different traffic models.

- The closed-form expressions of the system E2EPLP with the presence or absence of the network coding are obtained on the basis of the packet loss probability of D2D links. We find that both the number of relays and network coding have impacts on the system E2EPLP, end-to-end completion time, and average end-to-end transmission rate, which are critical performance indicators of the system.

The rest of this paper is organized as follows. Section II introduces the system model of multi-hop D2D communications underlaying the cellular network, and the associated channel models and traffic models. The D2D transmission with network coding and its packet loss probability are investigated in Section III. The general closed-form expressions of E2EPLP with or without network coding and the related performance indicators are given in Section IV. The simulation results validating these performance indicators are provided in Section V. Section VI concludes the paper. Also, notations used in the paper are summarized in Table I.

### II. SYSTEM MODEL

#### A. Network model

As shown in Fig. 1, we consider a general scenario of multi-hop D2D communications underlaying cellular networks. In the cell shown in Fig. 1, there are some randomly distributed cellular user equipment (CUE) and a pair of D2D user equipment (DUE1 and DUE2). These two D2D users need to exchange some multimedia contents like pictures, live videos, or interactive games. It is assumed that the distance between DUE1 and DUE2 is too long for DUE1 and DUE2 to directly communicate with each other using a one-hop D2D

<table>
<thead>
<tr>
<th>Notation</th>
<th>Meaning</th>
</tr>
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<tbody>
<tr>
<td>$M$</td>
<td>The number of relay nodes</td>
</tr>
<tr>
<td>$N_1, N_2$</td>
<td>The number of source packets correctly received by nodes $M+1$ and 0, respectively.</td>
</tr>
<tr>
<td>$K$</td>
<td>The Rician factor</td>
</tr>
<tr>
<td>$\lambda_{i,j}$</td>
<td>The transmission link from node $i$ to node $j$</td>
</tr>
<tr>
<td>$\lambda_j$</td>
<td>The average packet arrival rate at node $j$</td>
</tr>
<tr>
<td>$\mu_j$</td>
<td>The average service rate at node $j$</td>
</tr>
<tr>
<td>$\Delta T$</td>
<td>The fixed duration of each time slot</td>
</tr>
<tr>
<td>$A_j(\Delta T)$</td>
<td>The cumulative arrival amount at node $j$ during each time slot</td>
</tr>
<tr>
<td>$Z_j$</td>
<td>The maximum queue capacity of node $j$ in terms of packets</td>
</tr>
<tr>
<td>$Q_j(t)$</td>
<td>The queue length of node $j$ at the beginning of $t$-th time slot</td>
</tr>
<tr>
<td>$\gamma_0$</td>
<td>The SINR threshold at any node</td>
</tr>
<tr>
<td>$\sigma^2$</td>
<td>Additive white Gaussian noise</td>
</tr>
<tr>
<td>$\theta$</td>
<td>A variable in dB, $\theta = \frac{P_{T_i} d_{i,j}^{-\alpha} \Omega}{T_{M+2} d_{M+2,j}^{-\alpha} \omega + \delta^2}$</td>
</tr>
<tr>
<td>$T^T$</td>
<td>The total time slots with network coding</td>
</tr>
<tr>
<td>$T^E_{2E}$</td>
<td>The end-to-end completion time with network coding</td>
</tr>
<tr>
<td>$R^E_{2E}$</td>
<td>The average end-to-end transmission rate with network coding</td>
</tr>
<tr>
<td>$P_{rt,i,j}$</td>
<td>The packet loss probability of link $(i,j)$</td>
</tr>
<tr>
<td>$P_{bt,i,j}$</td>
<td>The packet delivery probability over link $(i,j)$</td>
</tr>
<tr>
<td>$P_{ft,i,j}$</td>
<td>The packet failure probability. $P_{ft,i,j} = 1 - P_{bt,i,j}$</td>
</tr>
<tr>
<td>$P_{cnt}$</td>
<td>The probability that node $i$ can encode packets</td>
</tr>
<tr>
<td>$P_{l,c}^{l_1,i,j}$</td>
<td>The packet loss probability of link $(i,j)$ with network coding</td>
</tr>
<tr>
<td>$P_{l,c}^{l_1,i,j}$</td>
<td>The end-to-end packet loss probability with network coding</td>
</tr>
<tr>
<td>$P_{l,c}^{l_2,i,j}$</td>
<td>The end-to-end packet loss probability without network coding</td>
</tr>
</tbody>
</table>
link. Therefore, some relay user equipment (RUE) need to be added to establish multi-hop D2D links between DUE\(_1\) and DUE\(_2\). We assume that the number of RUE is \(M\) and all of these relay users can be treated essentially as D2D users. The scheduling and resource allocation for all D2D user equipment (DUE\(_1\), DUE\(_2\) and RUE) can be done by the base station eNodeB. Also, it is assumed that all DUE reuses the same cellular uplink channel that is assigned to the \(i\)-th cellular user equipment (CUE\(_i\)). Moreover, the base station eNodeB has the control over the transmission power of the cellular user CUE\(_1\) and of all D2D user equipment to reduce the interference among them.

In Fig. 1, there are \(M + 3\) user equipment including one cellular user equipment and \(M + 2\) D2D user equipment. For simplicity, we name DUE\(_1\) as node 0, RUE\(_k\) as node \(k\) (\(k \in \{1, 2, \ldots, M\}\)), DUE\(_2\) as node \(M + 1\), and CUE\(_1\) as node \(M + 2\). The notation \((i, j)\) (\(i, j \in \{0, 1, \ldots, M + 1\}\), \(i = j \pm 1\)) is used to represent the transmission link from node \(i\) to node \(j\).

### B. Channel model

When node 0, node \(M + 1\), and relay nodes in Fig. 1 reuse the same cellular uplink channel that is assigned to node \(M + 2\), they can be subject to the interfering signals from node \(M + 2\). A strong interference from node \(M + 2\) may result in frequent interruptions of the D2D links, which will have a serious impact on the successful reception rate of the desired signals at D2D receiving nodes. Thus, the path loss and additive white Gaussian noise are introduced in our channel model. The large-scale fading is determined by the Euclidean distance \(d_{i,j}\) between node \(i\) and node \(j\) together with path loss exponent \(\alpha\), which is relevant to communication environments. The small-scale fading of link \((i, j)\) is characterized by a Rayleigh or Rician random variable \(f_{i,j}\). Therefore, the instantaneous signal-to-interference-plus-noise ratio (SINR) at node \(j\) with respect to node \(i\) and node \(M + 2\) is given by

\[
\gamma_j = \frac{P_{T_i} \cdot d_{i,j}^{-\alpha} \cdot F_{i,j}}{P_{T_{M+2}} \cdot d_{M+2,j}^{-\alpha} \cdot F_{M+2,j} + \sigma^2} = \frac{b \cdot c \cdot F_{M+2,j} + b \cdot \sigma^2}{b \cdot c}.
\]

where \(P_{T_i}\) is the transmit power of node \(i\), \(d_{i,j}^{-\alpha}\) is the path loss of link \((i, j)\) and \(F_{i,j} = |f_{i,j}|^2\) is the channel gain of link \((i, j)\). Similarly, \(P_{T_{M+2}}\) is the interference power of node \(M + 2\) with respect to node \(j\). The notations \(d_{M+2,j}^{-\alpha}\) and \(F_{M+2,j} = |f_{M+2,j}|^2\) are the path loss and channel gain of link \((M + 2, j)\), respectively. It is assumed that all nodes suffer the same additive white Gaussian noise \(\sigma^2\). For convenience, we let \(b = \frac{1}{P_{T_i} \cdot d_{i,j}}\) and \(c = P_{T_{M+2}} \cdot d_{M+2,j}^{-\alpha}\) in (1).

Moreover, node 0, node \(M + 1\), and relay nodes that form multiple D2D links are usually close next to each other, and this leads to the existence of dominant LoS signal components in the received desired signals. Also, these nodes are typically far away from node \(M + 2\), which results in no LoS signal components in the interference. Thus, in this paper, the Rician fading model is used for the channel gain of D2D links, and the Rayleigh fading model is used to deal with the interference links. We call such a setup Rician-Rayleigh channel fading model.

### C. Traffic model

If the network traffic is not controlled properly, network congestions will occur and subsequently compromise the network throughput, resulting in an inferior network performance. Multi-hop D2D communication system should fully consider the traffic characteristics at the relay nodes. Therefore, the packet inter-arrival time distribution, packet length distribution, traffic burst, behavioral characteristics of short- or long-term traffic, and QoS requirements of various services should be taken into consideration, in order to more accurately analyze the entire multi-hop D2D communication system.

As illustrated in Fig. 1, each node has certain queue capacity and service capacity. We assume that node \(j\) has a maximum queue capacity \(Z_j\) (packets) to dynamically store arrival data packets, and each packet has the same length of \(l\) (bits). Moreover, the transmission of data over the time is slot-by-slot based and each slot has a fixed duration \(\Delta T\) (ms). In each time slot, the spectrum resource can be allocated to one or more D2D links, depending on the resource sharing and scheduling strategies. During each time slot, the average service rate at node \(j\), \(\mu_j\) (packets/ms), is upper bounded by the channel capacity of the corresponding D2D link. With network coding, a certain number of data packets have to be buffered for the decoding purpose, and such a number is smaller than queue capacity and is negligible in our analysis. In the following, we will discuss several typical network traffic models.

The Pareto traffic model is a self-similar model that mainly describes the traffic inter-arrival time with a heavy-tailed probability density function (pdf). The inter-arrival time of such traffic is independent and identically distributed. Specifically,
the pdf of the inter-arrival time is the Pareto distribution given as follows [30]

$$f(x) = \frac{S \cdot \beta^S}{x^{S+1}}, \quad (x \geq \beta) \quad (2)$$

where $S$ is the shape parameter and $\beta$ is the minimum value of inter-arrival time. The mean and variance are respectively $\frac{S \cdot \beta^{S+1}}{S-1}$ and $\frac{S \cdot \beta^{S+1}}{(S-2) \cdot (S-1)}$, $1 < S < 2$. A larger value of $S$ indicates a higher self-similarity level which can lead to a higher packet loss probability and a more serious deterioration of network performance.

The Gaussian traffic model is another option that works well with the traffic input process. Due to the huge capacity of network links, hundreds or even thousands of multimedia contents are likely to be served by a multiplexer. Therefore, even though the traffic from each individual application cannot be characterized as a Gaussian process, by appealing to the central limit theorem, the aggregate traffic to the receiver can be effectively modeled as a Gaussian process [31].

The fractional Brownian motion (fBm) traffic model is an important self-similar model that primarily describes the cumulative amount of traffic arrivals and models the variation of connectionless traffic with a self-similar Gaussian process. A standard fBm random process $Y(t)$ with Hurst parameter $H \in (0.5, 1)$ is essentially a Gaussian process with the zero mean and the variance of $|t|^H$. As such, during each time slot, the cumulative amount of packets at node $j$, $A_j(\Delta T)$, which satisfies the self-similar fBm input traffic model, can be expressed as

$$A_j(\Delta T) = \lambda_j \cdot \Delta T + \sqrt{\frac{\eta \cdot \lambda_j \cdot Y(\Delta T)}{1}} \quad (3)$$

where $\eta$ is a variance coefficient, and $\lambda_j$ is the average packet arrival rate at node $j$ [32].

The Poisson traffic model, unlike other traffic models, is a conspicuous memoryless model (i.e., the future traffic behavior has nothing to do with the past traffic behavior). This model facilitates the traffic analysis process but cannot effectively reflect the burst nature of network traffic. During each time slot, the Poisson process of the cumulative amount of arrivals $A_j(\Delta T)$ at node $j$ with the parameter $\lambda_j$ is given by

$$\Pr\{A_j(\Delta T) = n\} = e^{-\lambda_j \cdot \Delta T} \left(\frac{\lambda_j \cdot \Delta T}{n!}\right)^n \quad (4)$$

where the expectation of $A_j(\Delta T)$ is equal to $\lambda_j \cdot \Delta T$ (i.e., $E(A_j(\Delta T)) = \lambda_j \cdot \Delta T$). Meanwhile, the packet inter-arrival time follows an exponential distribution with mean $1/\lambda_j$. With a proper selection of parameters this model can be used to simulate most of the network traffic reasonably well for a small duration of time.

Many studies on network traffic analysis have shown that in real-world networks, network traffic when investigated over a long period of time typically possesses some features such as burst, self-similarity, and long-range dependence [33]. Consequently, when the network carries a large number of data packets from various sources, the self-similar traffic models such as Pareto and fBm traffic models would be appropriate; otherwise, the Poisson traffic model that does not show a clear traffic burst would be proper.

D. The relay node selection and positioning

Another critical issue in multi-hop D2D communications is the relay node selection strategy (see Fig. 1). We assume that all relay nodes are able to and willing to relay data packets in our communication scenario, and participate in the relay node selection by using their local information. As shown in Fig. 2, the polar coordinate system is used in selecting relay nodes. (The blue dashed lines in Fig. 2 are for the proof associated with this figure in Appendix.) The pole is the position of the base station eNodeB, and the polar axis is the horizontal ray going through eNodeB and pointing to east. In such a setting, the polar coordinate of any node $j$ would be $(r_j, \theta_j)$. It is assumed that each node is equipped with a GPS system and the eNodeB has the knowledge of the location information of node $0$ and node $M + 1$ initially. $M$ relay nodes are selected in the following way: each relay node must be on the straight line connecting node $0$ and node $M + 1$, and the polar coordinates $(r_k, \theta_k)$ of the (ideal) relay node $k$ ($1 \leq k \leq M$) are given by

$$r_k = \frac{r_0 \cdot r_{M+1}}{r_0 \cdot \sin \theta_e + \frac{k \cdot \beta_0}{M+1} + \frac{k \cdot \beta_{M+1}}{M+1}}$$

$$\theta_k = \frac{M + 1 - k \cdot \theta_0 + k \cdot \theta_{M+1}}{M + 1} \quad (5)$$

where $\theta_e = \theta_{M+1} - \theta_0$. (A proof of (5) can be found in the Appendix of this paper.) Note that this relay node selection strategy can be characterized as the “shortest path” in the following sense. Assume that there are two candidate relay nodes $m$ and $m'$ where $m$ is on the straight line connecting node $0$ and node $M + 1$, but $m'$ is not. As such, by the laws of elementary geometry, the sum of the distance from node $0$ to node $m'$ and the distance from node $m'$ to node $M + 1$ will be longer than the sum of the distance from node $0$ to node $m$ and the distance from node $m$ to node $M + 1$. Therefore, the node on the straight line gives rise to the shortest possible distance. Since all relay nodes in our work are picked up on the straight line connecting node $0$ and node $M + 1$, we label such a strategy of choosing relay nodes as the “shortest path.”

III. D2D TRANSMISSION WITH NETWORK CODING AND ITS PACKET LOSS PROBABILITY

A. D2D transmission with network coding

In this section, we investigate the situation that the data packets are transmitted via relay nodes with network coding.
In multi-hop D2D links. When nodes 0 and M + 1 need to exchange some multimedia contents with each other, the network coding mechanism [34], called XOR coding, can be used to improve the transmission efficiency. Note that the majority of the studies on network coding assisted transmissions only consider the special case of one relay node (or two-hop D2D links), whereas a generalized scenario of M ≥ 1 relay nodes is tackled in our work. For the purpose of illustration, a special case of M = 2 (i.e., two relay nodes) is depicted in Fig. 3.

In Fig. 3, node 0 and node 3 need to transmit source packets (pkt1 and pkt2) to each other via relay nodes 1 and 2, and node 2 (in Fig. 3(b)) is the relay where the XOR coding operation is designated to take place. Fig. 3(a) depicts the scenario where no XOR coding is used; clearly, 6 time slots are needed in this case to have pkt1 and pkt2 exchanged between node 0 and node 3. Fig. 3(b), on the other hand, illustrates the situation where XOR coding is used to assist the transmission. In Fig. 3(b), node 2, after receiving pkt1 from node 0 (forwarded by node 1) and pkt2 from node 3, XORS the two and broadcasts the coded packet pkt1 ⊕ pkt2 to node 1 and node 3. Node 3 obtains the needed pkt1 by decoding the received coded packet; node 1 obtains the desired pkt2 in a similar fashion and then forwards it to node 0. As such, only 5 total time slots are needed to complete the exchange of the packets pkt1 and pkt2 between node 0 and node 3. Based on this, it is easy to see that in the general case of M relay nodes, the number of required time slots is 2(M + 1) with XOR coding, and 2(M + 1) without XOR coding.

In order to reduce the complexity in using XOR coding, we mainly perform one-to-one encoding of source packets from node 0 and node M + 1 through relay nodes. More specifically, suppose node 0 needs to transmit source packets (pkt_{t1}, pkt_{t2}, pkt_{t3}) to node M + 1, for example, and node M + 1 needs to send source packets (pkt_{s1}, pkt_{s2}, pkt_{s3}) to node 0. Then, pkt_{t1} is only XORed with pkt_{s1}, pkt_{t2} is only XORed with pkt_{s2}, and so on. When node 0 needs to send more source packets than node M + 1, the remaining source packets of node 0 (e.g. pkt_{t3}) will no longer be XORed with source packets from node M + 1. Instead, they will be transmitted directly in the multi-hop D2D links without being coded. Here, we assume that the D2D links have no packet loss. As such, when node M + 1 and node 0 respectively receive N1 and N2 source packets, we can deduce that the total number of time slots with the presence of XOR coding is

$$T^c_s = (N_1 + N_2) \cdot (M + 1).$$  \hspace{1cm} (6)

Similarly, the total number of time slots without using XOR coding is

$$T_s = (N_1 + N_2) \cdot (M + 1).$$  \hspace{1cm} (7)

### B. Packet loss probability

In Fig. 1, when node 0, node M + 1, and the relay nodes which form the D2D links chain reuse the same cellular uplink channel, these nodes will be subject to the interference from node M + 2. As mentioned before, this may lower the successful reception rate of desired signals at the receiving nodes since frequent D2D communication outages may occur due to the strong interference from cellular communications. On the other hand, note that network traffic also has an impact on the D2D data transmissions. Since the queue capacity and service capability of each relay node are limited, when the traffic amount exceeds a certain level, these relay nodes may start to malfunction causing network congestions and compromising the packet delivery. Therefore, in order to jointly deal with the impact from the cellular interference and network traffic conditions, we define the packet loss probability of D2D links as proposed in our previous work [22]:

$$Pr_{i,j}^l = Pr_{i,j} \cdot Pr_{i,j}^a = (1 - Pr_{i,j}) \cdot Pr_{i,j}^a,$$  \hspace{1cm} (8)

where $Pr_{i,j}^l$ is the packet loss probability of link (i, j), $Pr_{i,j}$ is the packet delivery probability defined to reflect traffic conditions over link (i, j), $Pr_{i,j}$ is the packet failure probability, and $Pr_{i,j}^a$ is the outage probability of link (i, j).

Note that Eq. (8) is carefully constructed to model the packet loss probability of D2D links from two perspectives – network layer and physical layer. While $Pr_{i,j}^l$ is a physical layer parameter that is determined by the channel condition and interference level, $Pr_{i,j}$ is a network layer parameter that indicates the network congestion caused by the traffic. It is reasonable to assume that $Pr_{i,j}^a$ and $Pr_{i,j}$ are independent of each other. In the following section, the closed-form expressions of $Pr_{i,j}^a$ and $Pr_{i,j}$ are respectively obtained using the Rician-Rayleigh channel fading model and various traffic models.

1) Link outage probability: The outage probability of transmission links needs to be studied on the basis of the relevant channel fading models. Most of the prior studies on D2D communications treat both the desired signals and the interference signals at the receiving nodes from the perspective of Rayleigh fading. This, however, is inaccurate since in close-range D2D communications a dominant part of the desired signals captured at the receiving nodes is actually the LoS signal. Thus, we use the Rician fading model in computing the the channel gain of D2D links, and the Rayleigh fading

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model in dealing with the interference signals received at the D2D nodes.

In (1), given that the envelope of the random variable \( f_{i,j} \) has a Rician distribution, the pdf of the random variable \( F_{i,j} \) has the exception value \( \Omega \). Also, note that the envelope of the random variable \( f_{M+2,j} \) has a Rayleigh distribution, which gives rise to its power being exponentially distributed. The pdf of the random variable \( F_{M+2,j} \) has the exception value \( \omega \).

In the Rician-Rayleigh channel fading model, the link outage probability is defined as the probability that the instantaneous SINR at node \( j \) in (1) is less than the SINR threshold \( \gamma_0 \), as follows \( \Pr_{i,j}^o = \Pr (\gamma < \gamma_0) \). Therefore, the link outage probability under the Rician-Rayleigh fading model can be expressed as

\[
\Pr_{i,j}^o = \frac{(K + 1) \gamma_0}{\theta + (K + 1) \gamma_0} \left[ 1 - \left( 1 - \frac{\sigma^2}{\theta} \right) \right] + \frac{\sigma^2}{\theta} \cdot \exp \left( -\frac{K}{\theta + (K + 1) \gamma_0} \right),
\]

where \( \theta = \frac{b \cdot \sigma}{b + c \cdot \sigma} \) is a variable in dB, \( K \) is the Rician factor defined as the ratio of the power of the LoS signal components to that of the scattered signal components. Note that Eq. (9) can also express the link outage probability under the Rayleigh-Rayleigh fading model when \( K = 0 \).

The proposed Rician-Rayleigh fading model might not be accurate when there are LoS signal components between the interfering node and the receiving node. However, a proper frequency channel assignment should avoid such a scenario. When the mean interfering power \( c \cdot \omega \) is much larger than white Gaussian power \( \sigma^2 \), Eq. (9) can be rewritten as

\[
\Pr_{i,j}^o \approx \frac{(K + 1) \gamma_0}{\theta + (K + 1) \gamma_0} \exp \left( -\frac{K \cdot \theta}{\theta + (K + 1) \gamma_0} \right).
\]

It is worth noting that this closed-form approximation is quite similar to the result of Yao and Sheikh [[35], Eq. (7)].

2) Packet failure probabilities: The packet failure probability is a network layer parameter that indicates the network congestion status caused by the traffic. When the queue of relay nodes reaches its capacity, the subsequent arriving packets will be dropped, which can lead to packet delivery failure. Let \( Q_j(t) \) be the queue length (packets) of node \( j \) at the beginning of the \( t \)-th time slot, which can be regarded as either a continuous-time or a discrete-time queue length. Here, \( Q_j(t) \) is considered to be a continuous-time queue length because equivalent results can be obtained for the discrete-time queue length. At the beginning of the \( (t + 1) \)-th time slot, the queue length \( Q_j(t + 1) \), the cumulative amount of arrival packets placed in the queue capacity, evolves as follows

\[
Q_j(t + 1) = \max \{ 0, Q_j(t) - \mu_j \cdot \Delta T + A_j (\Delta T) \}.
\]

It is assumed that each node needs to send a signal for requesting data forwarding to the base station before each time slot, that the arrived source packets during each time slot can only be transmitted at the next time slot, and that the base station archives the sequence of the data forwarding requests. When the queue length of node \( j \) exceeds its maximum capacity \( Z_j, Q_j (\Delta T) = Q_j (t + 1) - Q_j (t) \geq Z_j \), the packet delivery would fail. During the \( t \)-th time slot, the packet failure probability over link \((i, j)\), \( \Pr_{i,j} \), is defined as

\[
\Pr_{i,j} = \Pr \{ A_j (\Delta T) \geq Z_j + \mu_j \cdot \Delta T \}.
\]

Next, we derive the closed-form expressions of \( \Pr_{i,j} \) with different traffic models.

The fBm traffic model formalizes the variation of connectionless traffic with a self-similar Gaussian process. By substituting (3) into (12), we see that the packet failure probability over link \((i, j)\) in fBm traffic model is

\[
\Pr_{i,j} = 1 - \Phi \left( \frac{(\mu_j - \lambda_j) \cdot \Delta T + Z_j}{\sqrt{\eta \cdot \lambda_j \cdot \Delta T^H}} \right),
\]

where \( \Phi (x) = \int_{-\infty}^{x} \varphi (y) dy \) is the cumulative distribution function of the standard Gaussian distribution, and \( \varphi (x) = \frac{1}{\sqrt{2\pi}} e^{-\frac{x^2}{2}} \) is the probability density function. Note that \( \Phi (x) \) can be similar to the formula \( 1 - x^{-1} \varphi (x) \). We obtain the logarithmically asymptotical expression

\[
\Pr_{i,j} \approx \frac{\sqrt{\eta \cdot \lambda_j \cdot \Delta T^H}}{2 \pi} e^{-\frac{(\eta \cdot \lambda_j \cdot \Delta T^H + Z_j)}{2 \pi}}.
\]

Similarly, substituting (4) into (12) gives us the packet failure probability in the Poisson traffic model

\[
\Pr_{i,j} = 1 - e^{-\lambda_j \cdot \Delta T}, \sum_{n=0}^{\infty} \frac{\lambda_j \cdot \Delta T^n}{n!}.
\]

Since the Pareto traffic model is centered around the packet inter-arrival time, we cannot obtain an effective expression of cumulative amount of arrival packets. Fortunately, a Pareto/M/1 queueing model [36] through the use of the Pareto Laplace transform is studied. The packet failure probability under this queueing model is

\[
\Pr_{i,j} = \left[ 1 - \frac{S(S-1)}{\rho} X^{-\frac{S}{S-1} - 1} e^{X/2} \left( \sqrt{X} \cdot W_{-\frac{S}{S-1} - \frac{S}{2}} (X) - W_{-\frac{S}{2}} (X) \right) \right] V^{Z_j},
\]

where \( \rho = \frac{\lambda_j}{\mu_j} \) is the service utilization rate at node \( j \), \( W (\cdot) \) is the Whittaker’s function, \( X \) is equal to \( \left( \frac{S-1}{S} \right) \cdot \left( 1 - V \right) \) and \( V = S \cdot X^{-\frac{S}{S-1} - \frac{S}{2}} \cdot e^{-X/2} W_{-\frac{S}{S-1} - \frac{S}{2}} (X) \) is a geometric parameter.

IV. E2EPLP AND RELATED PERFORMANCE INDICATORS

In the previous section, we analyzed the packet loss probability of D2D links from a local perspective. In this section, we analyze the performance of the entire system. The general closed-form expressions of the end-to-end packet loss probability with or without XOR coding on the basis of \( \Pr_{i,j} \) in (8) are obtained. Then, the end-to-end completion time, the coding gain, and the average end-to-end transmission rate, are respectively investigated.
A. End-to-end packet loss probability

In Fig. 3(b), the probability that node 2 can successfully encode packets sent from node 0 and node 3, in time slot 3, depends on the receiving condition of these packets at each of relay nodes. Only after successfully receiving the packets from node 1 (which has to successfully receive the packets from node 0) and node 3, can node 2 perform the encoding process. In our work, without loss of generality, we assume that the packets encoding is done at the last relay node. As such, the probability that relay node $M$ can successfully encode packets is

$$ Pr^c_M = Pr^s_{M+1,M} \prod_{i=1}^{M} Pr^s_{i-1,i}, \quad (M \geq 1), $$

(17)

where $Pr^s_{i,j}$ is the packet success probability of link $(i, j)$. Consequently, the probability that node $M+1$ fails to successfully receive the source packets from node 0 when node $M$ can encode the source packets is

$$ Pr^l_{0,M+1} = 1 - Pr^c_M \cdot Pr^s_{M,M+1}. $$

(18)

Analogously, the packet loss probability from node $M+1$ to node 0 is

$$ Pr^l_{M+1,0} = 1 - Pr^c_M \cdot \prod_{j=1}^{M} Pr^s_{M+1-j,M-j}. $$

(19)

When node $M+1$ correctly receives $N_1$ source packets, the total amount of source packets that node 0 needs to transmit is $\frac{N_1}{1 - Pr^l_{0,M+1}}$. Also, when node 0 correctly receives $N_2$ source packets, the total amount of source packets sent from node $M+1$ to node 0 is $\frac{N_2}{1 - Pr^l_{0,M+1}}$. Therefore, the system E2EPLP with the presence of XOR coding is

$$ Pr^l_{E2E} = 1 - \frac{N_1 + N_2}{1 - Pr^l_{0,M+1}} + \frac{N_2}{1 - Pr^l_{0,M+1}}, $$

(20)

which can be rewritten as follows by substituting (18) and (19) into (20)

$$ Pr^l_{E2E} = 1 - (N_1 + N_2) \prod_{i=0}^{M} (1 - Pr^l_{i,i+1}) \times \frac{\prod_{j=0}^{M} (1 - Pr^l_{M+1-j,M-j})}{\prod_{j=1}^{M} (1 - Pr^l_{M+1-j,M-j})} N_1 + (1 - Pr^l_{0,M+1}) N_2. $$

(21)

Then, the system E2EPLP in the absence of XOR coding can be given by

$$ Pr^l_{E2E} = 1 - (N_1 + N_2) \frac{(1 - Pr^l_{0,M+1}) (1 - Pr^l_{M+1,0})}{1 - Pr^l_{M+1,0}} N_1 + (1 - Pr^l_{0,M+1}) N_2, $$

(22)

where $Pr^l_{0,M+1} = 1 - \prod_{i=0}^{M} (1 - Pr^l_{i,i+1})$ is the packet loss probability from node 0 to node $M+1$. $Pr^l_{M+1,0} = 1 - \prod_{j=0}^{M} (1 - Pr^l_{j,j+1})$ is the packet loss probability from node $M+1$ to node 0. In other words,

$$ Pr^l_{E2E} = 1 - \left( N_1 + N_2 \prod_{i=0}^{M} (1 - Pr^l_{i,i+1}) \right) \times \left( \prod_{j=0}^{M} (1 - Pr^l_{M+1-j,M-j}) \right). $$

(23)

By examining Eqs. (21) and (23), we can see that the mechanism of XOR coding and the number of relays have impacts on the system E2EPLP. Since the encoding/decoding operations of XOR coding occur at some relay nodes, the probability of these operations being successfully/unsucessfully performed has to be an extra part of the computation of system E2EPLP when the coding scheme is present. Moreover, considering that the packet loss process on each single-hop D2D link can be regarded as an independent Bernoulli event, an increased number of D2D links will inevitably increase the system E2EPLP. Similarly, a larger number of relay nodes would yield a higher system E2EPLP.

B. Other related performance indicators

In Section III, we saw that the total required number of time slots for the receiving node to correctly receive source packets is $T^c_s$. Note that it was computed under the assumption that there is no packet loss during the transmission, which means that the number of received source packets at the destination node is equal to the number of source packets sent at the source node. Here, we revisit the transmission time of source packets taking the packet loss probability into consideration. Using the aforementioned E2EPLP, the end-to-end completion time with XOR coding, $T^c_{E2E}$, which indicates the total transmission time required for correctly decoding packets at the destination node, can be calculated as

$$ T^c_{E2E} = \frac{T^c_s \cdot \Delta T}{1 - Pr^l_{E2E}}. $$

(24)

In a similar fashion, the end-to-end completion time without XOR coding can be formulated as $T_{E2E} = \frac{T^c_s \cdot \Delta T}{1 - Pr^l_{E2E}}$. Note that the computation time required for encoding and decoding the source packets during the transmission is ignored in the above calculations.

The notion of coding gain [34] is defined as the ratio of the total required packet transmission time without the presence of coding to that with the presence of coding (for the same set of packets). Thus, for the XOR coding, the coding gain would be

$$ G = \frac{T^c_{E2E}}{T^c_s}, $$

(25)

which reflects the benefits of XOR coding. Specifically, a larger the value of coding gain indicates a stronger benefit.

In addition, a new performance indicator, average end-to-end transmission rate, is introduced in [37]. In our XOR
coding, when node 0 correctly receives $N_2$ source packets, and node $M + 1$ correctly receives $N_1$ source packets, the corresponding average end-to-end transmission rate can be defined as

$$R_{E2E}^c = \frac{N_1 + N_2}{T_{E2E}}$$  \hspace{1cm} (26)$$

which captures the number of packets transmitted per time unit over the multi-hop D2D links.

V. SIMULATION RESULT

In this section, we conduct extensive simulations to analyze the performance of the proposed multi-hop D2D communication model with respect to the variation of several parameters. All parameters used in the simulations are shown in Table II.

Fig. 4 shows the D2D link outage probability in different channel fading models. Without loss of generality, we randomly select two SINR thresholds $\gamma_0 = -10$ dB and $\gamma_0 = -5$ dB in the simulations. In close-range D2D communications, a dominant part of the desired signals captured at the D2D receiving nodes is actually the LoS signal. Therefore, the Rician-Rayleigh channel fading model should be used. Here, the Rician factor $K$ is set to be 7 dB. Note that a larger Rician factor indicates stronger LoS signal components, and that $K = 0$ in fact entails that the link outage probability under the Rayleigh-Rayleigh channel fading model is to be computed. Clearly, we can see from Fig. 4 that for any fixed SINR threshold the link outage probability decreases as the $K$ increases, that a larger threshold leads to a higher link outage probability, and that the Rician-Rayleigh channel fading model which takes the LoS signal components into account yields a lower link outage probability than the Rayleigh-Rayleigh channel fading model. In order to keep the link outage probability low and facilitate the subsequent analyses, $\gamma_0$ is set to be -10 dB for the rest of the simulations.

Fig. 5 illustrates the packet failure probability in Pareto, fBm, and Poisson traffic models, and with respect to the queue capacity $Z$ and node service utilization rate $\rho$. While the queue capacity reflects the relay nodes’ ability to dynamically store arrival packets, the service utilization rate is defined to be the ratio of the packet arrival rate $\lambda_j$ to the service rate $\mu_j$. When the queue capacity is full or $\lambda_j > \mu_j$, the arriving packets are more likely to be discarded. As it can be seen in Fig. 5, a low queue capacity is more likely to result in a packet delivery failure, and $\rho > 1$ ($\lambda_j > \mu_j$) tends to cause more packet delivery failures than $\rho < 1$. To facilitate the subsequent analyses, $Z$ is randomly chosen to be 40 for the rest of the simulations. Moreover, it is worth noting that the packet failure probabilities of the Pareto and fBm traffic models, which take the burst and the self-similarity of network traffic as part of their consideration, are higher than that of the Poisson traffic model.

In a similar fashion, Fig. 6 depicts the packet loss probabil-
Fig. 6: Packet loss probability of D2D links in different traffic models.

Fig. 7: Packet loss probability of D2D links in Poisson traffic model.

Fig. 8: System E2EPLP with or without XOR coding in Poisson traffic model.

In Fig. 7, which shows the packet loss probability in Poisson traffic model indexed by $\theta$ and queue capacity $Z$, and by $\theta$ and service utilization rate $\rho$. We see that packet loss probabilities are higher when $\theta \in [-5, 0]$ and $Z \in [0, 50]$ (Fig. 7(a)), and when $\theta \in [-5, 0]$ and $\rho \in [1, 1.5]$ (Fig. 7(b)). For the sake of convenience of comparison, we assume that the packet loss probabilities of all D2D links between node 0 and node $M + 1$ in Fig. 1 are the same.

Fig. 8 depicts the system E2EPLP with respect to the presence/absence of XOR coding and the number of relay nodes $M$. Interestingly, we can see, in Fig. 8(a), that the mechanism of XOR coding actually raises the system E2EPLP. This should not be a surprise as XOR coding in fact increases the amount of information in the packets that are being transmitted, which, in turn, would result in an increase in the packet loss probability. Also, since the encoding/decoding operations of XOR coding occur at some relay nodes, the probability of these operations being successfully/unsuccessfully performed has to be (an extra) part of the computation of the system E2EPLP when the coding scheme is present. As for the number of relay nodes $M$, considering that the packet loss process on each single-hop D2D link can be regarded as an independent Bernoulli event, then, an increased number of D2D links (or of relays) will inevitably increase the system E2EPLP. On the other hand, an increased communication distance between the two D2D end users will certainly lead...
Fig. 9: System end-to-end completion time with and without XOR coding in Poisson traffic model.

Fig. 10: Coding gain in Poisson traffic model.

end-to-end packet loss probability, which, in turn, will lower the probability of packet being successfully decoded at the receiving end, and eventually increase the system end-to-end completion time. In a similar fashion, Fig. 10 shows the system coding gain in Poisson traffic model with different number of relay nodes. Since the end-to-end completion time with the presence of XOR coding is always less than that without XOR coding, the system coding gain by its definition will be always greater than 1.

Fig. 11 exhibits the system average end-to-end transmission rates with the presence/absence of XOR coding and with the number of relay nodes, when $N_1$ and $N_2$ are fixed. The result that the system average end-to-end transmission rate with the presence of XOR coding is greater than that with the absence of XOR coding, as shown in Fig. 11, can be attributed to the fact that the mechanism of XOR coding reduces the end-to-end completion time of the system. Another result from the Fig. 11 is that an increase in the number of relay nodes will result in a decrease in the average end-to-end transmission rate. This is actually what is expected, as a larger number of relays will yield a longer end-to-end completion time.

Fig. 12 (a) and (b) depict the comparison of our work (XOR coding) with RLNC and batch coding [37] in terms of end-to-end completion time and average end-to-end transmission rate, respectively. Since the framework of the D2D communications in [37] is two-hop (i.e., one relay node), we set $M = 1$ in our work for a fair comparison. As shown in Fig. 12(a), the end-to-end completion time of our work is the highest one. This is not surprising as encoding of packets is conducted in a batch mode in both RLNC and batch coding schemes (i.e., either $N_1$ source packets are encoded at node 0 or multiple packets buffered at the relay nodes are encoded and then forwarded), but only in a one-to-one fashion in our work. This fact allows both RLNC and batch coding use much less time slots than the XOR coding, resulting in a much less end-to-end completion time than our work, as shown in Fig. 12(a). Fig. 12(b) can be understood in a similar way since the end-to-end transmission rate is inversely proportional to the end-to-end completion time.

Note that although the performance of XOR coding is worse
than that of the other two coding approaches in terms of end-to-end completion time and end-to-end transmission rate, the implementation of XOR coding is much simpler than that of the other two approaches. This simplicity is actually preferred when energy efficiency is much of a concern in D2D communications in which all mobile devices are subject to energy constraints.

VI. CONCLUSION

We have proposed a general network coding-assisted multi-hop D2D communication system underlaying the cellular network, together with its associated relay node selection strategy, in which the base station determines the position of all relay nodes by using the location information of the D2D pair and in conformity with the criterion of the “shortest path”. In the proposed system, both Rician-Rayleigh channel fading model and Rayleigh-Rayleigh channel fading model are investigated. It turns out that the Rician-Rayleigh channel fading model, which takes the LoS signal components into consideration, outperforms the Rayleigh-Rayleigh channel fading model in terms of D2D link outage probability. Also, the packet delivery status in different traffic models were thoroughly analyzed. We have found that Pareto and fBm traffic models, which include traffic properties such as burst and self-similarity as an essential part in their formations, yield a higher packet failure probability than that of the Poisson traffic model. In addition, the system E2EPLP with the presence or absence of network coding were computed using the packet loss probability of D2D links, and we have noticed that the network coding scheme and the number of relays both can impact the performance of the system. In particular, three critical system performance indicators, namely, E2EPLP, end-to-end completion time, and average end-to-end transmission rate are all subject to the presence/absence of the network coding scheme and the change in the number of relays.

APPENDIX A

PROOF OF FORMULA (5)

In Fig. 2, $x$ and $y$ are the two legs of the right triangle formed by eNodeB, node $k$, and the line segment that goes through node $k$ and is perpendicular to the line connecting eNodeB and node 0.

Note that

$$
\tan \angle Node0 = \frac{x}{r_0 - y} \quad \text{and} \quad \tan (k\Delta \theta_e) = \frac{x}{y},
$$

from which can obtain

$$
y = r_0 \cdot \tan \angle Node0 = r_0 \cdot \tan (k\Delta \theta_e) + \tan \angle Node0.
$$

Thus,

$$
r_k = \frac{y}{\cos (k\Delta \theta_e)} = \frac{r_0 \cdot \tan \angle Node0}{\sin (k\Delta \theta_e) + \tan \angle Node0 \cdot \cos (k\Delta \theta_e)}.
$$

Note also that

$$
\tan \angle Node0 = \frac{r_{M+1} \cdot \sin \theta_e}{r_0 - r_{M+1} \cdot \cos \theta_e},
$$

so the polar coordinates $(r_k, \theta_k)$ of relay node $k$ are

$$
r_k = \frac{r_0 \cdot \sin \frac{k\theta_e}{M+1} + r_{M+1} \cdot \sin \frac{(M+1-k)\theta_e}{M+1}}{M+1},
\theta_k = \theta_0 + \frac{k}{M+1} \theta_e.
$$

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