

# Optimal Configuration of Intercommunicate Coding in Adhoc Network

Vadde.Saikavya<sup>1</sup>&,Gogineni Jyothi<sup>2</sup>& Mr. B. Laxmaiah<sup>3</sup>

<sup>1</sup>M-Tech,Dept of CSE, Sarada Institute of Science and Technology (SITS), Khammam.

<sup>2</sup>Associate Professor, Dept of CSE, Sarada Institute of Science and Technology (SITS), Khammam.

<sup>3</sup>HOD,Associate professor, Dept of CSE, Sarada Institute of Science and Technology (SITS), Khammam.

**Abstract**—Network coding has recently emerged as an effective solution for multicast and broadcast communications in ad hoc networks. We focus on broadcast traffic and design a network coding-based scheme that we compare against simpler solutions, by using the network simulator ns-2. Indeed, while often the benefits of network coding have been shown via theoretical analysis or in simplified simulation scenarios, our aim is to assess the performance of network coding in ad hoc networks when a realistic MAC protocol is considered. Our results show that the performance of network coding for traffic broadcasting strongly depends on the network node density and on the generation size. In particular, network coding leads to significant gains in terms of end-to-end packet loss probability and protocol overhead only when the average number of neighbors per node is higher than.

**Index Terms**—Ad hoc networks, traffic broadcasting, network coding

## 1. INTRODUCTION

Recent results on the advantages of network coding in wired networks have stimulated a lot of interest in the subject and in particular, in the application of network coding to wireless ad hoc networks. Network coding refers to the basic notion of performing coding operations on the contents of packets throughout a network, and is generally attributed to Ahlswede et al. [1], who showed the utility of the network coding

for multicast in wired networks. The work of Ahlswede et al. was followed by other work by Koetter and Medard [5] who showed that codes with a simple, linear structure were sufficient to achieve the capacity of multicast connections in lossless, wireline networks. This result was augmented by Ho et al. [3], who showed that, in fact, a random construction of the linear codes was sufficient. The utility of such random linear codes for reliable communication over

lossy packet networks such as wireless ad hoc networks was soon realized [7]. In [8], a prescription for the efficient operation of ad hoc networks is given, which proposes using the random linear coding scheme of [7] coupled with optimization methods for selecting the times and locations for injecting coded packets into the network. In ad hoc networks, broadcasting is used for data dissemination with several different aims such as finding a route to a particular node, sending a warning signal or performing service discovery. The simplest method to implement broadcasting is flooding, where every node in the network retransmits a message to its neighbors upon receiving it for the first time. Flooding, however, can lead to undesired effects such as the well-known broadcast storm problem: redundant packet retransmissions resulting in repeated contention, collisions, and extra power consumption. Two typical ways to alleviate the broadcast storm effects consist in reducing possible redundant transmissions and differentiating the timing of retransmissions, as extensively discussed in [1].

## 2. RELATED WORK

### Existing System

The probability that the random linear NC was valid for a multicast connection

problem on an arbitrary network with independent sources was at least  $(1 - d/q)\eta$ , where  $\eta$  was the number of links with associated random coefficients,  $d$  was the number of receivers, and  $q$  was the size of Galois field  $Fq$ . It was obvious that a large  $q$  was required to guarantee that the system with RLNC was valid. When considering the given two factors, the traditional definition of throughput in ad hoc networks is no longer appropriate since it does not consider the bits of NC coefficients and the linearly correlated packets that do not carry any valuable data. Instead, the good put and the delay/good put tradeoff are investigated in this paper, which only take into account the successfully decoded data. Moreover, if we treat the data size of each packet, the generation size (the number of packets that are combined by NC as a group), and the NC coefficient Galois field as the configuration of NC, it is necessary to find the scaling laws of the optimal configuration for a given network model and transmission scheme.

### Disadvantages

- Throughput loss.
- The decoding loss.
- Time delay.

### Proposed System:

Proposed system with the basic idea of NC and the scaling laws of throughput loss and decoding loss. Furthermore, some useful concepts and parameters are listed. Finally, we give the definitions of some network performance metrics. Physical layer Network Coding designed based on the channel state information (CSI) and network topology. The PNC is appropriate for the static networks since the CSI and network topology are preknown in the static case. There are  $G$  nodes in one cell, and node  $i$  ( $i = 1, 2, \dots, G$ ) holds packet  $x_i$ . All of the  $G$  packets are independent, and they belong to the same unicast session. The packets are transmitted to a node  $i'$  in the next cell simultaneously.  $g_{ii'}$  is a complex number that represents the CSI between  $i$  and  $i'$  in the frequency domain.

#### Advantages:

- ✓ System minimizes data loss.
- ✓ System reduces time delay.

### 3. IMPLEMENTATION

#### Network Coding-based Broadcasting

Our network coding-based broadcasting is implemented on top of the network layer running over an IEEE 802.11 MAC protocol. This choice allows us to avoid encoding routing address information carried in IP headers. As suggested in [10], we embed the coding vector in the packet header, thus

dispensing with the need for centralized knowledge of the graph topology or decoding functions. In particular, the encoding header follows the IP header as the first piece of data in the IP payload. The network coding header contains information about the encoded packet, such as: the generation size and identifier, the number of encoded packets along with their size, and the coding vector. We now detail the different operations performed by the source node, the intermediate nodes, and the receiver nodes. It is worth remarking that, in broadcasting, intermediate nodes are also receiver nodes; however, for the sake of clarity, we distinguish the two operations. Also, note that each node (except for the source) needs to collect  $s$  (recall that  $s$  is the generation size) independent packets for further encoding/decoding, therefore some buffer space is needed at the receiver. In the following, we will call this buffer NC buffer.

#### Simple Flooding

The simple flooding scheme we use as term of comparison is based on the IEEE 802.11 standard. When a node sends a broadcast packet, all its one-hop neighbors will receive it. Since all the neighbors need to rebroadcast the packet in their turn, their transmissions may occur at almost the same time, thus

likely resulting in collisions. To reduce the collision probability, we introduced a simple link-layer modification to the standard: after receiving the broadcast packet, we let the receiving node defer the retransmission by a small time, randomly chosen in the range.

### Deferred Broadcast

The deferred broadcast scheme we consider includes the mechanism proposed in [5], which, for the sake of clarity, we briefly recall. Upon receiving a broadcast packet, nodes store the information about the predecessor node and, before taking a decision whether to rebroadcast it or not, they wait for a hold-off time  $T_{ho}$ , during which they listen to possible retransmissions of

the same packet by other nodes. In particular, for each packet, the node counts the number of rebroadcast events from other nodes with different predecessors. If this counter exceeds a given threshold, the packet is dropped. We set this threshold to 2, since simulation results show that a greater value does not lead to any improvement. This choice is supported by the findings in [1], where it is stated that if a node rebroadcasts a packet heard more than twice, the extra area it can cover is only about 9%.

## 4. EXPERIMENTAL RESULTS

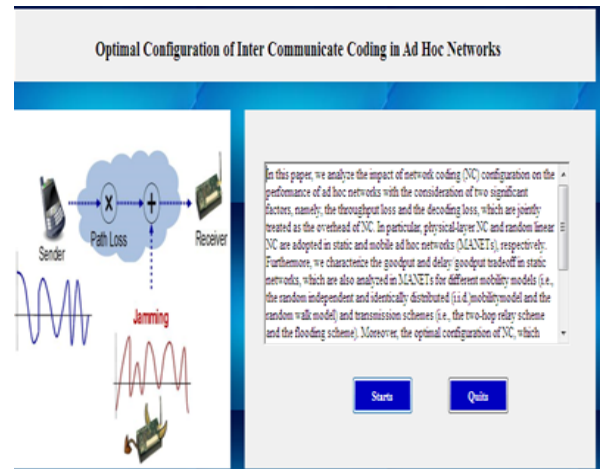


Fig:-1 Base Station

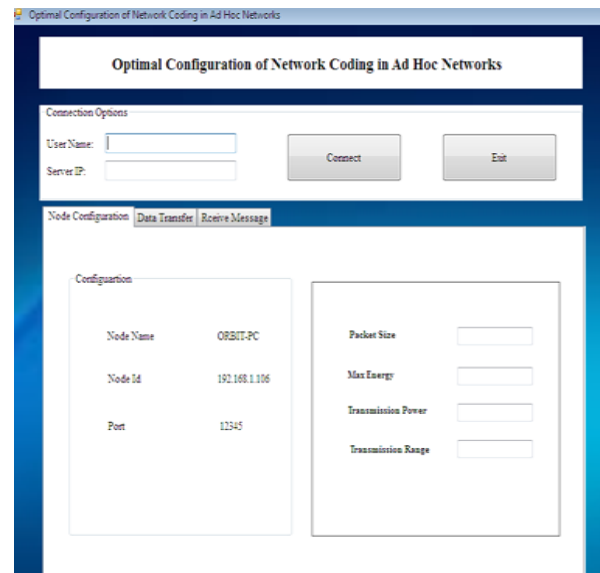


Fig:-2 Node Station

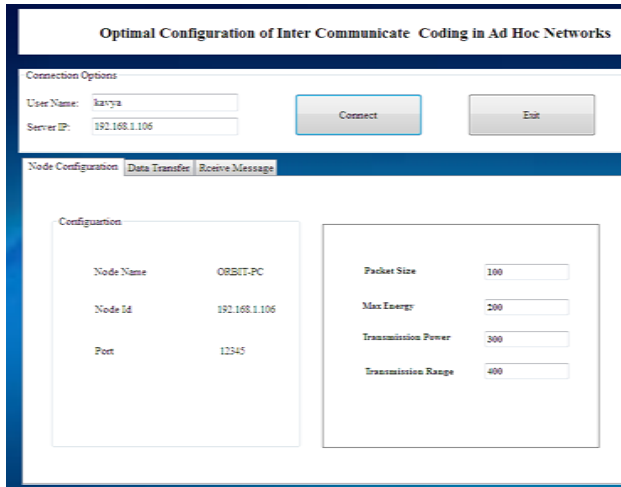


Fig:-4 Data Connection

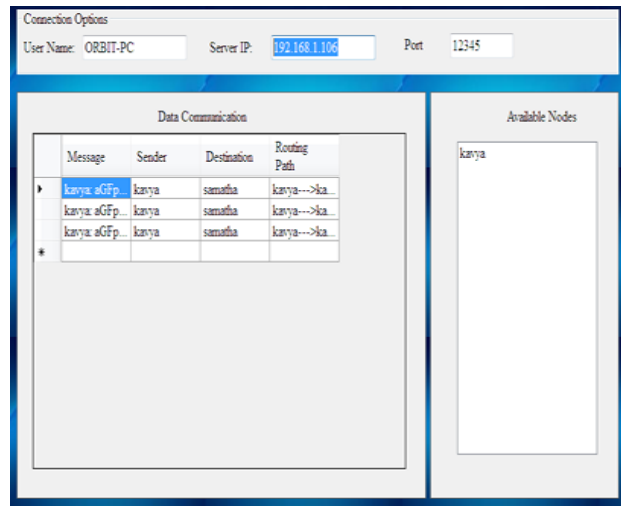


Fig:-5 Senders path

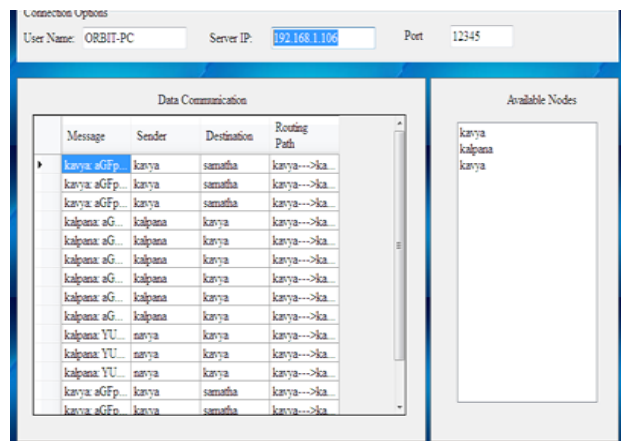


Fig:-6 Reviser End

## 5. CONCLUSIONS

In this paper, we discussed the problem of identifying the maximum throughput that a multicast connection with or without network coding can achieve in an unreliable static ad hoc network. We presented mathematical programming formulations for the problem that include wireless medium contention constraints, which are crucial to the problem, and, through numerical analysis using the formulations, we show that network coding achieves 65% higher throughput than conventional multicast in a typical setting. In addition, we showed through simulation that network coding allowed very robust communications with significantly less overhead than conventional multicast.

## 6. REFERENCES

- [1] R. Ahlswede, N. Cai, S.-Y. R. Li, and R. W. Yeung. Network information flow. *IEEE Trans. Inform. Theory*, 46(4):1204–1216, July 2000.
- [2] M. Goemans and Y. Myung. A catalog of steiner tree formulations. *Networks*, 23:19–28, 1993.
- [3] T. Ho, M. Médard, R. Koetter, D. R. Karger, M. Effros, J. Shi, and B. Leong. A random linear network coding approach to multicast. Submitted to *IEEE Trans. Inform. Theory*.

[4] K. Jain, J. Padhye, V. Padmanabhan, and L. Qiu. Impact of interference on multi-hop wireless network performance. In Proc. ACM MOBICOM, 2003.

[5] R. Koetter and M. Médard. An algebraic approach to network coding. IEEE/ACM Trans. Networking, 11(5):782–795, Oct. 2003.

[6] S.-J. Lee, M. Gerla, and C.-C. Chiang. On-demand multicasting routing protocol. In Proc. of IEEE WCNC '99.

[7] D. S. Lun, M. Médard, R. Koetter, and M. Effros. On coding for reliable communication over packet networks. Submitted to IEEE Trans. Inform. Theory.

[8] D. S. Lun, N. Ratnakar, M. Médard, R. Koetter, D. R. Karger, T. Ho, E. Ahmed, and F. Zhao. Minimum-cost multicast over coded packet networks. IEEE Trans. Inform. Theory, 52(6):2608–2623, June 2006.

[9] J.-S. Park, D. S. Lun, Y. Yi, M. Gerla, and M. Médard. Codecast: A network coding based ad hoc multicast protocol, 2006. Submitted to IEEE Wireless Communications.

[10] Qualnet. <http://www.scalable-networks.com>.

## Authors Profile

### Student



**Name :** VADDE.SAIKAVYA

B.tech in SREEKAVITHA ENGINEERING COLLEGE(SKEC), Khammam, percentage is 76.76% , year of completed April 2013.

M.tech[CSE] (Computer Science And Engineering)

**College:** Sarada Institute of Technology & Science(SITS), Khammam.

**Mail id:** [saikavya.592cse@gmail.com](mailto:saikavya.592cse@gmail.com)

### Guide

**Name:** Gogineni Jyothi.

**Experience:** 10 years.

**Qualification:** M.Tech from JNTU, Hyderabad.

**Designation:** Associate Professor.

**Working:** Sarada Institute of Technology & Science(SITS), Khammam

**Email-d:** [jgogineni@gmail.com](mailto:jgogineni@gmail.com)

### HOD



Technology & Science (SITS), Khammam. He obtained M.Tech degree from JNTUH, Hyderabad. His research areas include Object Oriented Programming Through Java, Data base Management System, Data Structures, Web Services, Data Warehousing and Data Mining and Operating Systems.

**Name:** Mr. B. Laxmaiah

**Working:** Head of the Department,  
Associate professor CSE, Sarada Institute of