hJam: Attachment Transmission in WLANs

Kaishun Wu, Member, IEEE, Haochao Li, Student Member, IEEE, Lu Wang, Student Member, IEEE, Youwen Yi, Student Member, IEEE, Yunhuai Liu, Member, IEEE, Dihu Chen, Member, IEEE, Xiaonan Luo, Member, IEEE, Qian Zhang, Fellow, IEEE, and Lionel M. Ni, Fellow, IEEE

Abstract—Effective coordination can dramatically reduce radio interference and avoid packet collisions for multistation wireless local area networks (WLANs). Coordination itself needs consume communication resource and thus competes with data transmission for the limited wireless radio resources. In traditional approaches, control frames and data packets are transmitted in an alternate manner, which brings a great deal of coordination overhead. In this paper, we propose a new communication model where the control frames can be "attached" to the data transmission. Thus, control messages and data traffic can be transmitted simultaneously and consequently the channel utilization can be improved significantly. We implement the idea in OFDM-based WLANs called *h*Jam, which fully explores the physical layer features of the OFDM modulation method and allows one data packet and a number of control messages to be transmitted together. *h*Jam is implemented on the GNU Radio testbed consisting of eight USRP2 nodes. We also conduct comprehensive simulations and the experimental results show that *h*Jam can improve the WLANs efficiency by up to 200 percent compared with the existing 802.11 family protocols.

Index Terms-Wireless network, interference, coordination, OFDM

1 INTRODUCTION

COORDINATION among stations can effectively reduce radio interference and avoid packet collisions in multistation wireless local area networks (WLANs). Coordination needs communication and stations have to exchange control messages to well coordinate. The control messages can be delivered in an explicit, implicit, or hybrid manner. However, all control messages will consume valuable communication resources such as the communication channel and transmission air time.

In a practical WLAN, the transmissions of control messages and data traffic often interleave. As illustrated in Fig. 1a, the current CSMA/CA protocols (e.g., 802.11 a/g/n) transmit the control messages and data traffic in an alternate manner. Between data traffic there are always fractions of air time for coordination purposes such as DIFS, SIFS, backoff, and packet acknowledgment. It is well known that such mechanism is quite inefficient when data frames are small [1]. When higher physical layer (PHY) data rates are

- K. Wu is with College of Computer Science and Software Engineering, Shenzhen University, and also with Guangzhou Fok Ying Tung Research Institute. E-mail: kwinson@ust.hk.
- H. Li, L. Wang, Y. Yi, Q. Zhang, and L.M. Ni are with the Department of Computer Science and Engineering, HKUST Fok Ying Tung Graduate School, Hong Kong University of Science and Technology, Hong Kong SAR, China. E-mail: {stevenli, wanglu}@ust.hk, euwen.17@gmail.com, {qianzh, ni}@cse.ust.hk.
- Y. Liu is with the Third Research Institute of Ministry of Public Security, China. E-mail: yunhuai.liu@gmail.com.
- D. Chen is with the School of Physics and Engineering, Sun Yat-sen University, Guangzhou 510006, China. E-mail: stscdh@mail.sysu.edu.cn.
- X. Luo is with the National Engineering Research Center of Digital Life, State-Province Joint Laboratory of Digital Home Interactive Applications, School of Information Science and Technology, Sun Yat-sen University, Guangzhou 510006, China. E-mail: lnslxn@mail.sysu.edu.cn.

Manuscript received 29 Nov. 2011; revised 23 Aug. 2012; accepted 30 Aug. 2012; published online 12 Sept. 2012.

supported, the efficiency becomes even worse because of the shortened data traffic air time. Off-the-shelf 802.11n products now support up to 300 Mbps PHY data rate, while the effective throughput is only 60 Mbps [1].

To deal with this issue, a direct way in traditional approaches is to separate the control messages and data traffic. In this approach (e.g., [22]), a dedicated PHY channel is allocated for coordination. This approach consumes an entire channel for control purposes only, which is also too expensive. The separation can also be done in other dimensions. Side Channel [15] transmits the control messages in the code space. It is a customized design for direct sequence spread spectrum (DSSS) modulation only and does not have general applicability.

Rather than interleaving or separating the control messages and data traffic, serving them together at the same time is more desirable. As illustrated in Fig. 1b, in this model the data traffic and the control messages are transmitted simultaneously in the same channel. Data traffic accounts for the entire fraction of transmission air time and is allocated the same bandwidth as in traditional systems. In the meanwhile, control messages are transmitted in an attached manner with the data traffic. As such the coordination overhead can be dramatically reduced.

This idea is simple but very challenging to realize. It is mainly because in the Fig. 1b scenario the control messages and data traffic are transmitted from independent transmitters. These transmitters will have no extra coordination and thus are very likely to collide with each other. It becomes even more challenging when there are several control messages from different transmitters. In a typical WLAN, it is common that when one node is transmitting the data, all others may have the demands to transmit their requests.

Recently, interference cancellation (IC) technique [5], [14], [30] has been developed well which brings a new hope. Since a practical rate adaptation scheme is unlikely to operate at the ideal bitrate, there will always be a slack for

For information on obtaining reprints of this article, please send e-mail to: tmc@computer.org, and reference IEEECS Log Number TMC-2011-11-0639. Digital Object Identifier no. 10.1109/TMC.2012.194.



(b) Desired communications - hJam



IC to exploit. By a successful application of this technique we propose a new communication architecture called hJam with the core idea in Fig. 1b. hJam is built on top of orthogonal frequency division multiplexing (OFDM) networks, as OFDM has been widely adopted in modern WLAN protocols (e.g., 802.11 a/g/n) and is becoming the standard for the next generation of WLANs (e.g., WiMAX and 3GPP LTE).

*h*Jam enables two kinds of transmissions in communications. One is the *high-throughput transmission* for data traffic and different application data traffics compete for it. Highthroughput transmissions share the same (de)modulation method, (de)coding algorithm, and the bandwidth with the original OFDM system, so are fully compatible to the traditional OFDM systems. The other is an *attachment transmission* that allows each high-throughput transmission to carry a number of small-sized attachments. The attachments are independent to the high-throughput transmissions, and thus are extremely suitable for control message delivery. Attachment transmission supports multiple accesses though the high-throughput transmission supports one flow at a time.

*h*Jam is based on the observation that in current OFDM systems there is a packet preamble for channel estimation purpose. The preamble may have redundancy in different wireless link conditions due to correlation of the channel response of different subcarriers. When this redundancy is smartly utilized, a small amount of information can be delivered by intentionally injecting jamming signals. For control messages, this amount of information is sufficient. When implementing hJam we encounter many practical challenging issues. The first challenge is that in real environments the redundancy in preamble is marginal and dynamic due to the dynamic environment. Second, jamming signals are not necessarily generated intentionally, instead they could be the result of noise. We have to design smart jamming signal generators and detectors to reduce the miss detections and false alarms. Last, coordination in attachment transmissions is also a challenge.

To address these challenges, in hJam we design customized jamming signals so that they are not likely to coincide with the noise. An interference cancellation technique is applied to cancel the effects of the jamming signals. We also design adaption schemes to accommodate the various environments and leverage the OFDM feature to enable multiple accesses. In summary, the main contributions of this paper are as follows:

- We propose *h*Jam, a new PHY architecture for OFDM-based WLANs that enables concurrent transmission of coordination message and traffic data to improve the coordination efficiency. To the best of our knowledge, it is the first of its kind in the literature to enable simultaneous transmissions for control messages and traffic data in OFDM systems.
- We analyze the reliability of *h*Jam theoretically, and numerical results show that multiple attached coordination information can be decoded correctly and the original data traffic is not affected with high probability.
- We demonstrate the feasibility of *h*Jam by implementing it on a GNU Radio testbed of 8 USRP2 nodes. We also simulate *h*Jam performance in a large scale network. *h*Jam shows significant higher efficiency than prior protocols that do not allow concurrent transmission of coordination and data. It can provide up to 2× gain in efficiency, as compared to traditional 802.11 standards.

The remainder of this paper is organized as follows: In Section 2, we give a brief overview of the OFDM. Then in Section 3 the system architecture is given. This is followed by the detail design of hJam in Section 4. In Section 5, we analyze the performance of hJam. The implementation of hJam is presented in Section 6. Experimental evaluations are given in Section 7. In Section 9, the related work is shown. Finally, conclusions are presented and suggestions are made for future research.

2 BACKGROUND OF OFDM

In this section, we introduce the background information of the OFDM system which is the foundation for *h*Jam design. OFDM is a digital multicarrier modulation method for wideband wireless communications. It is widely adopted by many wireless standards such as IEEE 802.11a/g/n, WiMAX [7], and as the core technique for future standards such as 3GPP LTE [8]. OFDM divides the spectrum band into many small and partially overlapped signal-carrying frequency bands named subcarriers as shown in Fig. 2. These subcarrier frequencies are carefully chosen so that the cross-talk between subcarriers sums up to zero. As a result, subcarriers are orthogonal to each other and can thus be packed tightly without putting guard bands between them. On the transmitter side, the data to be transmitted on an OFDM signal is spread across the carriers of the signal, each carrier taking part of the payload. This baseband modulation is performed via an inverse fast fourier transform (IFFT). To combat symbol misalignment (due to multipath effects), OFDM has a built-in robustness mechanism called cyclic-prefix (CP). Instead of using an empty guard space, a cyclic extension of the OFDM symbol fills the gap. Then, the signal sequence with CP is converted into analogue signals and transmitted to air.



Fig. 2. OFDM framework [25].

Upon receiving the signals, the receivers sample the signals and pass them to a demodulation process chain. After a sampling procedure, the data sample blocks will be processed by an FFT process and the final result is the original data subject to certain scaling and phase rotations. These scaling and phase rotations are mainly due to channel dispersion. Therefore, channel equalization is needed to recover the original data from the distorted one.

In OFDM-based WLANs, the time/frequency synchronization and channel estimation are performed using short and long preambles which are located in the header of each transmission packet. Due to channel correlation in the frequency domain, the channel responses of some subcarriers can be interpolated with those of the neighboring ones [28]. This property is exploited in our attached transmission design.

3 OVERALL SYSTEM ARCHITECTURE

In this section, we present the system architecture of *h*Jam which allows simultaneous transmissions for both control messages and data traffic. Challenges in the system design are also presented in this section.

3.1 hJam Communication Paradigm

The hJam PHY architecture introduces several new components. At the transmitter end, a jamming generator is



Fig. 3. An illustrative example of hJam communication system.



Fig. 4. Architecture of hJam communication system.

designed to enable attachment transmission when necessary. At the receiver end, we introduce a jamming detector to detect the jamming signals, an attachment analyzer to decode the attachment transmission, and an interference cancellation engine to cancel the effects of attachment transmission and recover any high-throughput content.

Consider a simple transmission scenario with four clients Alice, Bob, Carol, and Dave, and an AP, as illustrated in Fig. 3, and the architecture of *h*Jam communication system is depicted in Fig. 4.

Suppose Alice obtains a high-throughput channel for the next transmission. Alice is in normal mode which transmits the content in the traditional way (High-throughput Tx. in Fig. 4 is exactly the same as OFDM Tx.). The others (Bob, Carol, and Dave) will then turn to the hJam mode and attempt to use the attachment transmission. Each client in hJam mode will select a unique subcarrier assigned by AP and send jamming signals when Alice is sending. These jamming signals carry the attached information from the hJam clients, combine Alice's signal in the air and are received by the AP. At the receiver end, the AP first applies the jamming detector to determine whether any jamming signals from hJam clients exist. These jamming signals are then analyzed and decoded to recover attachment transmissions (from Bob, Carol or Dave). In the meantime, the interference cancellation technique [5] is applied to cancel the jamming signals and recover the original data for the high-throughput client (Alice).

3.2 Design Challenges

The design principles of *h*Jam are simple and effective. In practice, however, the implementation of *h*Jam design faces many practical challenges.



Fig. 5. Jamming effect on one subcarrier.

First, the success of *h*Jam depends highly on whether the generated jamming signals can be reliably identified and decoded. Note that we may not necessarily intend to generate the jamming signals, instead they could be the result of noise. We have to carefully design the jamming signal generator and detector, striking a tradeoff between miss detections and false alarms and strive to reduce both.

Second, the coordination information is attached on the data traffic. The receiver should be able to decode both kinds of data. However, in current mature OFDM-based WLANs, with proper rate adaptation and channel coding, the BER performance is sensitive to the interference. Therefore, it is a great challenge to detect both attached information and original data.

Last, the bandwidth for attached transmission is limited. Therefore, it is important to modulate multiple coordination messages effectively, and coordinate them to avoid collision. Otherwise collisions in the attachment channel may cause the interference cancellation to fail, resulting in a failed high-throughput channel transmission which may have severe consequence.

In the next section, we give details on how we address these challenges in *h*Jam design.

4 HJAM DESIGN

In this section, we detail the design of hJam communication architecture. hJam introduces several new components, namely a jamming generator on the transmitter side, and a jamming detector, interference cancellation and attachment analyzer on the receiver side. We describe the design of each component in detail.

4.1 Jamming Generator and Detector

For each hJam mode client (Bob, etc., in Fig. 3), it needs to encode its control messages and transmit them through the attachment transmissions by using our designated jamming generator. As mentioned in Section 3.1, each client in hJam will be assigned with a unique subcarrier. To guarantee that different hJam clients do not interfere with each other's jamming signals, we intentionally narrow the jamming signal channel width so that it is completely inclusive of a single subcarrier even in presence of frequency offset. Fig. 5 depicts an illustrative example of narrowed jamming signals (dash-line) within a subcarrier.

Accordingly, at the AP side, a jamming detector is carefully designed to identify these jamming signals from

Jam	ming detector for k-th subcarrier
1:	Initialize P_{avg} and P_{std} ;
2:	$P_{avg} = RSS(s_1)$, $P_{std} = RSS(s_1)$;
3:	For each transmission with n symbols $\{s_1, s_2 \dots s_n\}$;
4:	For each $s_i, i \in [1, n]$
5:	if $RSS(S_i) > P_{avg} + \lambda \cdot P_{std}$;
6:	return true;
7:	else $P_{avg} = \alpha \dot{P}_{avg} + (1 - \alpha) \cdot RSS(s_i);$
8:	$P_{std} = \sqrt{\alpha \dot{P}_{std} + (1 - \alpha) \cdot (RSS(s_i) - P_{avg})^2}$
9:	End if
9:	End for
10:	End for;
11:	return false

Fig. 6. Pseudocode of the jamming signal detection algorithm for each of the subcarriers.

the noise. Specifically in our current design, we adopt a simple yet effective scheme by using energy detection. This is based on the simple observation that in general cases, high-throughput transmissions and noise have an even energy distribution over the spectrum. When there is a signal burst existing in a subcarrier (i.e., the combined signal strength of both the data and jamming signal in our case), it is very likely this is due to intended behavior. More detailed system implementations of the energy detection will be introduced in Section 6.

As long as such intentional jamming signals are successfully detected, we are able to cancel corruption effects induced by jamming and recover the highthroughput channel data by leveraging our interference cancellation component.

Fig. 6 gives the pseudocode of the energy-based jamming signal detection algorithm. We omit a detailed explanation because of the simplicity.

4.2 Interference Cancellation

The main objective of interference cancellation is to cancel the jamming signals and recover the content in the highthroughput channel. Notice that the raw signal is not directly decodable as it combines both high-throughput transmissions and the attachments.

As mentioned in Section 2, due to channel correlation, the channel estimation of some subcarriers can be interpolated with neighboring ones and thus it is sufficient to use only part of the subcarriers to send pilot in preambles. We call the rest vacant subcarriers as clean symbols because ideally no signal except noise is received at these subcarriers. In our attachment transmission design, we exploit this opportunity and make use of those clean subcarriers to record the jamming signal for the purpose of recovering data signal.

In wireless communication, the received signal is typically represented as a stream of discrete complex symbols spaced by the sampling interval T. These symbols are different from the transmitted ones both in amplitude and phase. For example, the data signal x[t] without jamming on the *i*th subcarrier can be expressed as

$$y_i[t] = H_i x[t] + w[t],$$
 (1)

where $H_i = he^{j\omega}$ is a complex number, the magnitude h refers to channel attenuation and the angle $j\omega$ is a phase shift that depends on the distance between the transmitter and the receiver, and w[n] is a random complex noise.



Fig. 7. Example illustration of attached information.

More specifically in our design, when jamming signals are introduced in those subcarriers not used for sending preambles, the received signal in clean subcarrier can be expressed as

$$y_i''[t] = y_B[t] + w[t], (2)$$

where $y_B[t] = H_B x_B[t]$ refers to the jammer's signals after traversing their corresponding channels to the receiver. Accordingly, the received signal in those data symbol combines the data signal and the jamming signal and can be further expressed as

$$y_i'[t] = y_A[t] + y_B[t] + w[t],$$
(3)

where $y_A[t] = H_A x_A[t]$ refers to the transmitter's signals after traversing their corresponding channels to the receiver. Thus, the original data signal can be recovered by canceling the jamming signal from the received signal in the data symbol. Specifically, it is by making use of (2) and (3) as follows:

$$\hat{x}_A[t] = \frac{y_i'[t] - y_i''[t]}{H_A},\tag{4}$$

where H_A can be further estimated by the training sequence.

4.3 Modulation/Demodulation of Attached Information

With the above techniques we are able to correctly identify individual jamming signals that are intentionally generated from an *h*Jam node. In this section, we see how to modulate and demodulate the attachment to such jamming signals.

Different from a traditional decoder, we trade the jamming signal as intended information rather than noise. With synchronization, we can decide whether to jam the data at a specific subcarrier for one symbol duration time or not. For the ease of decoding, we use a jamming signal in each symbol to represent one bit of information, i.e., the jammed subcarrier in the symbol is considered to be "1" and the clean one to be "0," or otherwise. Since one packet is modulated over symbols, and each symbol is modulated on subcarriers, these jamming patterns can be represented by a bit sequence, which will be the transmitted attachment. As illustrated in Fig. 7, the attached information in subcarrier 2 is "01010011001." Notice that the jamming signals for the attachment transmission starts from the first

data symbol in the packet, and ends with the last symbol of the same packet. The jamming signal in the preamble is used by interference cancelation for recovering the original data. Therefore, in such a design, the capacity of attachment transmission mainly depends on the following three factors: the number of data symbols per packet (i.e., n), the number of jamming nodes (i.e., m) and the time for the normal packets transmission (i.e., t). Thus, it can be further expressed as $n \cdot m/t$. The value m is further bounded by the number of total available subcarriers for jamming in the system, which is equal to half of the number of total subcarriers in the communication system, the capacity of the attachment channel is bounded by $n \cdot N/2t$.

Obviously, the attached information able to transmitted by a specific user is bounded by the number of data symbols per packet *n*. However, the number of subcarriers for jamming is not limited by the total number of OFDM subcarriers, but the performance degradation of the original data link that can be tolerated due to the existence of these jamming signals. In the next section, we analyze the effect of the jamming signals on performance of the highthroughput transmission.

4.4 Multiple Access by hJam

To demonstrate the effectiveness of hJam in this section, we show how to use hJam to benefit transmissions in WLANs in the infrastructure mode [21]. The high-throughput transmission is used for application data traffic and the attachment is used for control message delivery.

Consider the single AP scenario. Upon receiving a data packet from a client, AP first decodes both the highthroughput and attachment transmissions. Then the highthroughput content is delivered to the upper layer application directly while the attachments are collected for coordination purpose. These attachments carry the transmission requests from the clients and can be further used to build a potential sender list. By having this list, the AP is responsible for whole channel coordination and assign the next sender. Specifically, the AP attaches the senders' IDs in order in the ACK and broadcast it. At the client end, by receiving the ACK from AP, the client can check its order in the sender list and determines whether it is the next sender of the high throughput channel. Then the next data transmission continues. It will be similar when the transmission is from the AP to the client due to that each client knows its sending order.

In addition, clients may join and leave. At the initialization step, the AP is responsible for allocating the subcarriers to the existing clients in the network. Afterward, a Client being inactive for too long time is automatically kicked out by the AP. To the contrary, a new comer should first listen to the AP's broadcast ACK packet (indeed, the ACK is for other clients). This packet carries the subchannel utilization information and the new comer simply selects a random unused subcarrier to delivers its request.

5 PERFORMANCE ANALYSIS

In this part, we analyze the performance of *h*Jam. The first issue is to find out the conditions under which the



Fig. 8. Relationship between BER, PRR, # of clients and SNR(AWGN with noise and jamming, SIR = 17 dB).

attachment transmission is nearly harmless to the performance of original data transmission in terms of packet reception rate (PRR), so that data traffic can be guaranteed under *h*Jam. The second issue is to evaluate the performance of attachment transmission in terms of jamming detection rate (JDR), so that control messages can also be guaranteed. To this end, a key parameter N_u , which is the maximum number of subcarriers we can totally jam, is derived for different channel conditions.

Except the data transmission, the performance of the attachment transmission itself should also be evaluated in terms of the JDR, so that the probability of missing a jamming signal when one is present (miss detection), and the probability of falsely detecting a jamming signal when it is absent (false alarm), is designed to be as small as possible. We mainly focus on these two factors in the following sections.

5.1 Performance for Data Symbol

The first influential factor used to measure the quality of original data transmission is PRR. According to [25], we can depict the relationship between PER and bit error rate (BER) in Fig. 8, where left axis is BER and right axis is the corresponding PRR. For example, if we require PRR to exceed 99.6 percent, then the desired BER should be less than 10^{-5} .

BER has a direct connection with the encoding/decoding scheme applied by original data transmission. To be safe, the joint effect of intended jamming and noise should not go beyond the error correction capability of that coding/ decoding scheme. Here we adopt convolutional encoder as the channel coding scheme and accordingly Viterbi hard decision decoder as the channel decoding scheme.

Lemma: For a hard decision, the Viterbi algorithm is a minimum Hamming distance decoder. An upper bound on the BER is used to examine its performance [25]:

$$P_b = \frac{1}{k} \sum_{d=d_{free}}^{d_{free}} B_d P_d, \tag{5}$$

 P_d is the probability of selecting a code word what is Hamming distance *d* from the correct word. When *d* is even:

TABLE 1 Notations for BER Calculation

k/n	number of information/coded bits in convolu-			
	tional code			
r	r = k/n is defined as channel coding rate			
d	hamming distance			
d_{free}	free distance of the convolutional code			
B_d	total number of information bit ones on all weight			
	d paths			
P	the uncoded probability of bit error in AWGN			
	under the effect of jamming			
B_j/B_o	B_j/B_o bandwidth of jamming signal/OFDM symbol			
ρ	$\rho = N_u * B_j / B_o$ is the total jamming portion			
E_b/N_0	ratio of average energy per bit-to noise power			
	spectral density			
Na	jamming nower spectral density			

$$P_d = \sum_{i=\frac{d+1}{2}}^{d} {d \choose i} p^i (1-p)^{d-i}.$$
 (6)

When d is odd:

$$P_d = \frac{1}{2} {\binom{d}{\frac{d}{2}}} p^{\frac{d}{2}} (1-p)^{\frac{d}{2}} + \sum_{i=\frac{d+1}{2}}^d {\binom{d}{i}} p^i (1-p)^{d-i}.$$
 (7)

Table 1 lists the notations used for calculating P_b . Here, we consider different type of modulation schemes for a single OFDM subcarrier, including BPSK, QPSK, 16QAM, and 64QAM. for simplicity we only compute BPSK/QPSK, 16/64 QAM will be showed in Fig. 8 below. With the presence of noise and jamming in the original data transmission, p with BPSK/QPSK can be expressed as

$$p = \rho \cdot Q\left(\sqrt{\frac{2rE_b}{N_0 + N_j/\rho}}\right) + (1-\rho) \cdot Q\left(\sqrt{\frac{2rE_b}{N_0}}\right).$$
(8)

We depict (5) in Fig. 8, which shows the relationship between BER, PRR, and SNR when different number of attachment transmissions exist. The figure shows that when the channel condition exceeds a certain threshold (e.g., SNR > 10 dB), BER hardly changes as SNR increases. BPSK/QPSK shows robust performance even with N_u = 20 attachment transmissions, where hjam obtain a desired $BER < 10^{-9}$ and a corresponding PRR > 100%. While with 16/64QAM, BER increases as the number of concurrent transmission increases. In the range from 20 to 30 dB which is the typical working range of 802.11 [27], BER remains stable at 10^{-7} for the worst case (64QAM with 20 concurrent transmissions), resulting in a PRR > 99.7%. This is acceptable, and further confirms that the performance degradation induced by the attachment transmission can be ignored. Therefore, we can come to the conclusion that in theory *h*jam is harmless and can be safely used in WLANs.

5.2 Performance for Attachment Transmission

Now, we evaluate the performance of attachment transmission in terms of jamming detection rate, which is dominated by the probability of miss detection P_{miss} and false alarm P_{false} . According to our jamming detection algorithm, when the energy strength of received signal R(d) exceeds certain threshold (λ) , we determine the presence of a jamming signal at instant d.



Fig. 9. Relationship between SIR, P_{miss} , P_{false} and SNR ($N_u = 1$).

Lemma. Given a certain threshold value λ , P_{miss} , and P_{false} can be expressed as [26]:

$$P_{miss}(\lambda) = 0.5 erfc\left(\frac{\mu_M - \lambda}{\sqrt{2\sigma_M^2}}\right),\tag{9}$$

$$P_{flase}(\lambda) = e^{-\lambda G^2/D^2}.$$
 (10)

Here μ_M and σ_M are mean and variance of R(d) inside jamming signal, while D and G^2 are mean and variance of R(d) outside jamming signal. According to [26], the total SIR has a reverse impact on P_{miss} and P_{false} . We depict this feature in Fig. 9. As is shown below, P_{miss} decreases while P_{false} increases as the total SIR increases. Based on this observation, the total SIR should be set appropriately to meet the requirements of both P_{miss} and P_{false} . For example, if SNR is around 10 dB, we can get a miss detection rate of $P_{miss} < 10^{-45}$ and false alarm rate $P_{false} < 10^{-48}$ by setting SIR =15 dB, when the number of attachment transmission $N_u = 1$, which is small enough for 802.11 specifications.

Now we derive how to calculate N_u using (9) and (10), which are depicted in Fig. 10. When the channel condition is above 10 dB, we can set $N_u = 20$ to obtain a desired miss detection rate of $P_{miss} < 10^{-25}$ and false alarm rate of $P_{false} < 10^{-18}$. This result also agrees with N_u calculation under BER factor. Taking P_{miss} and P_{false} together into consideration, we can evaluate the probability that successful detecting the whole packet of attachment transmissions P_h :

$$P_h = \left(\frac{1}{2}(1 - P_{miss}) + \frac{1}{2}(1 - P_{false})\right)^{P_L},\tag{11}$$

where P_L is the packet length of one attachment transmission, with $P_L = 40$ bits for example, the probability that correctly detecting an attachment transmission packet is 99.99 percent. Therefore, we can conclude that *h*Jam, in theory, is not only harmless but reliable in typical working range of 802.11.

6 System Implementation

Building an operational communication system, however, involves many practical challenges. We use GNU Radio testbed for our experiments. We have implemented *h*Jam



Fig. 10. Relationship between # of clients, P_{miss} , P_{false} , and SNR (SIR = 17 dB).

using software defined radios (SDRs). The SDRs are from the open source GNU Radio project [10], which implement signal processing blocks of wireless communication system in software. We use the Universal Software Radio Peripheral 2 (USRP2) [11] for our RF frontend, and use the RFX2450 daughterboard which operate in the 2.4-2.5 GHZ range. Our implementation uses BPSK as the modulation. We have implemented the basic mechanisms of *h*Jam on the USRP2.

One challenge during the implementation is the strict timing requirements due to synchronization (measured in microseconds). If the clients and AP are not synchronized, the misalignment between the jamming signal and data signal may lead to the failure of the interference cancellation. However, the unpredictable latency caused by signal processing in software makes precise time control impossible in GNU Radio and thus software radios are incapable of real synchronization. To compensate for this latency, we import the USRP2 timestamps derived from the radio hardware [16] to record the packet receiving time. Thus upon receiving the ACK, we are able to control the data/ jamming transmission time of all clients by adding a constant delay after the ACK's receiving time, so that all the senders can transmit the data at the same time.

Another challenge is the threshold setting for the jamming detection. According to our system design, either false alarm or missing detection will cause misbehavior in the interference cancellation procedure and thus lead to performance degradation. As the distance and the transmission power vary, how to dynamically adjust the threshold for accurate detecting jamming becomes the key issue in our system design. Currently, we have applied exponential averaging to track the mean (μ) and standard deviation (σ) of normal power level (w/o jamming) and set the threshold as $\mu + \lambda \cdot \sigma$. A peak exceeding this threshold is recognized as the jamming signal.

Finally, the last challenge is the spectrum leakage and the phase offset. During the experiment, we observed that the subcarrier signal often has some spectrum leakage. Also the phase offset will cause jamming misalignment. We have noted that as mentioned in [17], the channel width can be adaptive. We then narrowed down the subcarrier channel width as it becomes a portion of the original one.



Fig. 11. Experimental environment (three sets of the three nodes' locations are illustrated as an example).

7 EXPERIMENTAL EVALUATION

In this section, we first evaluate the feasibility of hJam by using our prototype implementation on an indoor testbed consisting of up to 8 USRP2 nodes. We then implement a custom simulator to study the performance of hJam over 802.11 under a large scale wireless network. The real-time experiment was conducted under a single AP network with a varying number of clients. Our experimental results show that the impact by adopting harnessing jamming on the normal packet transmission is negligible and the simulation results further suggest that hJam outperforms the traditional CSMA MAC of 802.11 in all cases and improves the system throughput by up to 200 percent at high physical data rates.

7.1 Experiment

The feasibility of *h*Jam is the focus of our experiments and mainly consists of two aspects. First, hJam is feasible only if the jamming signal can be cancelled out and thus the normal data packets can be recovered. Second, due to the coordination of the whole system design highly depending on the control message encoded in the jamming signals, the decodability of the attachment transmission becomes essential and dominates the effectiveness of hJam. Thus, we have conducted real-time experiments by using USRP2 nodes with RFX2450 daughterboards operating in the 802.11 frequency range in our office, which is a typical real-world environment with size $5 \text{ m} \times 8 \text{ m}$ shown in Fig. 11. Unless otherwise specified below, we use the default configuration, for example, a packet size 400 bytes. And we use BPSK as our modulation scheme. Specifically, we use 52 subcarriers and a bandwidth of around 2 MHz, resulting in a subcarrier spacing with around 38 KHz. We make these changes because we want to make the inter subcarrier spacing comparable to 802.11 (0.3125 MHz) while still maintaining the normal transmission of USRP2, which is limited by the hardware itself. All of our experiments run on the 2.425 GHz.

7.1.1 Is hJam Harmless?

To answer this question, experiments are conducted to examine the decodability of *h*Jam and the impact of the number of concurrent jamming clients, respectively.

For measuring the decodability, we use a three-node setting, i.e., a sender, an AP, and a jamming client. Upon receiving an ACK from AP for acknowledgment and



Fig. 12. Decodability of hJam under different SNRs.

coordination, the sender sends a normal packet while the jamming client sends the jamming signal simultaneously. Then we evaluate this by comparing the PRR under various SNRs. Each run transfers 2,500 packets, first without jamming, then with jamming. For each value of SNR, we repeat the experiment 10 times.

Fig. 12 plots the PRR with/without the jamming client as a function of the received SNR at AP ranging from [4, 20] dB. The figure shows that when the SNR exceeds a threshold, i.e., larger than 10 dB, the PRRs almost have no difference between the cases with and without a jamming client. This little performance degradation is acceptable because the typical range of SNR region defined for 802.11 is 10-30 dB [27], where our *h*Jam works well. Though, such results are comparable but less than the theoretical optimum. We infer this is due to two reasons. First, the software-defined signal processing may limit the USRP2's ability of the strict timing and accurately sampling. Second, our implementation runs in a public user-space in the unlicensed 2.4 GHz range, some external interferences cannot be avoided.

Though we have proved the feasibility of the *h*Jam's decodability in the presence of jamming signal, it is also interesting to ask whether the number of concurrent jamming clients has impact on the system performance. To investigate this influence, we use a similar setting to evaluate the PRR of the normal sender but with the number of concurrent jamming clients varying from 1 to 6. More precisely, we have 52 subcarriers (from 0 to 51) and those subcarriers with odd number are not used for channel estimation. The six subcarriers we have used for jamming in this experiment are Subcarrier 1, 3, 5, 7, 9, and 17.

Fig. 13 plots our results under different number of jamming clients with SNR 11 dB and 15 dB, respectively. We expected that the performance loss would increase when the number of the concurrent jamming clients increases as shown in our theoretical analysis (see Fig. 8). Surprisingly, we observe that the relative performance loss varies randomly under different number of concurrent jamming clients, though they are so small that can be negligible. In theoretical analysis, even when SNR is 11 dB with 20 jamming clients, the BER is below 10^{-6} , which would lead to merely no performance loss. We infer that the



Fig. 13. Impact of number of concurrent jamming clients under different SNRs.

difference between the practical and the theoretical results may due to the processing capability of USRP2 hardware.

7.1.2 Is the Attachment Transmission Reliable?

This question refers to the detection accuracy problem, i.e., whether we can accurately detect the jamming signal and decode the attached information correctly. It is mainly affected by the miss detection and false alarm rate. Note that hJam apply the interference cancellation only when it detects a jamming signal. So both cases will cause misbehavior in interference cancellation procedure and result in the decoding failure.

In this part of the experiment, we use similar three-node setting but varying the SNR of the jamming signal with a fixed SNR of data signal being around 15 dB. For each run, the sender sends 2,500 packets in total while the jamming client keeps sending attached information encoded in the jamming signal upon receiving an ACK. The AP logs all the attachment information for calculating the results. For each SNR value ranging from [8, 20] dB, we repeat the experiment 10 times.

From the experimental results, we find out that there is no false alarm which is consistent with the theoretical analysis. Fig. 14 shows that when SNR > 13 dB, the miss detection rate is controlled within 1 percent, resulting in a detection accuracy more than 99 percent. The detection algorithm will also impact the results. Therefore, one of our future works is to design a more precise detection algorithm.

7.2 Performance Evaluation

The latency constraint of USRP2 disallows the real-time evaluation of the system throughput. Thus, we implement a simulator to understand the performance of hJam primarily under a single AP network with varying number of total clients. To focus on the performance on the channel utilization by each method, we assume that the packet reception failure is only caused by the collisions and the network is saturated. In this section, we mainly compare the performance of hJam with CSMA/CA [12], which is used by the current IEEE 802.11 Standard.

Our simulations model the CSMA MAC. For scheduling of *h*Jam, we just simply use Round Robin as an example to



Fig. 14. Miss detection rate under different SNRs.

demonstrate its performance. For the evaluation of hJam, (12) gives a simple model for hJam's throughput efficiency

$$E_{hJam} = \frac{t_{data}}{t_{preamble} + t_{data} + t_{ACK}}.$$
 (12)

Unless otherwise stated, the default packet size is 1,500 bytes, which is around the maximal transmit unit (MTU). Table 2 summaries the configuration parameters used in our simulators. Specially, for the simulation of high data rates with 802.11n, the total number of subcarriers is set to 114, of which 108 are used for data transmission and 6 for equalization.

Fig. 15 plots the throughput gain of hJam over 802.11 as a function of total clients under different data rates. It shows that hJam outperform 802.11 CSMA MAC at all cases and its throughput gain is significant, for example, when the data rate is 54 Mbps, the relative throughput gain over 802.11 a/g is up to around 72 percent. Fig. 16 further illustrates the throughput gain over 802.11n in higher data rates. Compared to 802.11n at a data rate 600 Mbps, hJam can even achieve a gain up to 200 percent. This significant improvement is due to two reasons. First, hJam eliminates the coordination overhead in each transmission while the proportion of coordination overhead in CSMA increases as the data rate increases. The second reason is due to that hJam is collision free while the collision probability of CSMA in IEEE 802.11 increases with the number of clients. Thus hJam has better utilization of the channel.

7.3 Summary of the Results

Our experiments and simulations reveal the following:

TABLE 2 Configuration Parameters

Parameters	Values	Parameters	Values
SIFS	$10\mu s$	DIFS	$28\mu s$
PIFS	$19\mu s$	Slot time	9μs
Preamble	$20\mu s$	Symbol time	$4\mu s$
CW_{min}	16	CW_{max}	1024



Fig. 15. Performance gain of hJam over 802.11 a/g under different number of clients.

- We have prototyped the implementation of *h*Jam on a testbed consisting of eight USRP2 nodes to prove the feasibility of *h*Jam. *h*Jam provides a high decodablity around 100 percent and achieves a 99 percent detection accuracy when SNR is greater than 13 dB.
- *h*Jam provides significant throughput gain over tradition 802.11 CSMA MAC by up to 200 percent.

Moreover, the feasibility results in this paper are derived from laboratory experiments, without node mobility. *h*Jam under harsh conditions reminds for our future work.

8 DISCUSSION

Some practical issues in the *h*Jam are discussed in this section.

- Multiple AP scenarios. Besides the infrastructure 1. mode, here we consider a network with multiple APs in different collision domain. Actually as we mentioned before, we can solve this problem by assigning them orthogonal channels. However, if the channels are not enough and should be nonorthogonal, how to coordination the clients in the overlapped region becomes a challenge. In that case, we reserve one subcarrier for the contention of these clients. The clients in the overlapped region between two APs will contend the channel through the reserved subcarrier. And the contention procedure is similar to the traditional CSMA. Then the associate AP will also use this subcarrier to send back the ACK. Since we reserve one subcarrier for the clients in the overlapped region, the performance degradation is $1/48 \approx 2\%$. However, compare with the traditional CSMA, we still have significant throughput gain.
- 2. *Priority of transmission.* For some latency sensitive transmission, the priority of the transmissions will significantly affect the network performance. For example, SSH only needs to send several bits information but has to wait for a long time based on the traditional contention methods. Especially for some real-time applications, the delay is intolerable. However, in the traditional approaches, all nodes



Fig. 16. Performance gain of *h*Jam over 802.11n under different number of clients.

needs to contend the channel before transmission and the workload information of the data traffic is not taken into account for priority scheduling. In hJam, the load indicator [20] of the packets can be delivered as the attached information. Then, we can design some smart scheduling algorithms based on such information to reduce the network latency.

9 RELATED WORK

To address the radio interference issues and reduce the transmission collisions well, a large amount of coordination schemes have been proposed. The existing approaches can be classified as out-of-band and in-band.

The out-of-band coordination approaches are more suitable for multichannel/radio environments. In these approaches, they often allocate a dedicated PHY channel to control messages [22]. In out-of-band approaches, Stations switch in and off the control channel during transmissions, leading to significant switching overhead. In addition, these approaches consume an entire channel for control purpose only and the interference or congestions at the control channel will result in the waste of the whole bandwidth of data channel, which is too expensive. Recently, in the work [2], [3], it moves the channel contention from time domain to frequency domain. However, their work assumed that every station has two antennas, one for transmitting and the other for listening. In such approaches, one channel and antenna are wasted, providing no effective gain for the network throughput.

The in-band approaches deliver the control traffic in the same channel as the data traffic. It will also consume the communication resources. In the current 802.11 legacy protocol design [12], the coordination is scheduled along the temporal space, which introduces great overhead such as the DIFS, SIFS and random back-offs. Some recent work also reveal the need for optimal CSMA by experimental results [19]. In [13], Zeng et al. propose a minimum controlled coordination by reducing the DCF overhead. However, in our *h*Jam, we remove such coordination overhead.

In some recent works, researchers begin to think of improving the channel utilization by some PHY designs. In FICA [1], a fine grained channel access system is proposed. It has improved the channel utilization by increasing the data transmission time in each subchannel and thus increases the whole efficiency. Actually, from the (12) we can understand that there are two major factors to influence the channel utilization. One is the overhead and the other is data transmission time. If we want to improve the efficiency, we can either decrease the overhead or increase the data transmission time. FICA choose the second option while hJam use the first one. FICA effectively reduces the channel contention level but still needs DIFS, SIFS and random back-offs between transmissions. In hJam, we focus on the coordination overhead and tried to eliminate them. These two methods try to improve the channel efficiency in a different way. Actually, we believe these two methods can be combined together and thus significantly solve the low utilization problem in wireless networks. This will be left for our future work. Side Channel [15] transmits the control messages in the code space but works for DSSS modulation only. For the high data rate modulations such as OFDM, it is not applicable. To enable coordination information delivery in more general environments, we design hJam on the modern OFDM modulation schemes. In hJam, the information in the high-throughput channel is completely recoverable. This is mainly due to the successful application of the interference cancellation technique in the design. Recall that in Side Channel, the intended interference patterns are carefully tuned, which may otherwise ruin the original channel transmission. Halperin et al. [18] use the channel measurements (CSI) from commodity 802.11 NICs to improve the channel estimation. We plan to implement *h*Jam on the commercial products as part of our future work.

*h*Jam is also different from traditional TDMA and FDMA in terms of design purpose. *h*Jam is designed for efficient channel contention to solve the overhead problem in current 802.11 standard which uses CSMA, while TDMA and FDMA are designed for data transmission of multiple nodes in time and frequency, respectively. Moreover, TDMA and FDMA require very tight time synchronization and thus high complexity of the equipment, which is widely used in cellular network but not suitable to adopt in lowcost WiFi devices.

10 CONCLUSION

Coordination is a well-known problem in wireless networks that causes significant performance degradation. We find that fundamentally it is because of the interleaved control messages and data traffic account for too much of the transmission air time. Rather than separating or interleaving them, we propose a new communication model that transmit them together and develop hJam to realize it on top of OFDM-networks. Intended jamming signals and interference cancellation techniques are used with the explored preamble redundancy in OFDM modulations. As such, in hJam the application data can obtain the full air time. Theoretical analysis confirms the general applicability of hJam in practical environments. We implement hJam on GNU Radio testbed. The performance evaluations as well as the simulations show that hJam outperforms IEEE 802.11 protocols by up to 200 percent under different traffic patterns.

We will conduct the future work along the following directions. First, *h*Jam is in particularly designed for single AP scenario. In a multiple-AP scenario *h*Jam is applicable

but there will be other concerns such as the fairness issues. Second, the latency constraints with the USRP platform brings many limitations. Later we will use more powerful hardware (e.g., Sora [4]) to implement hJam to fully explore its capability. Third, the attachment transmission in hJam is used for control messages in this paper. Indeed, it can be utilized for more general purposes with low-latency and small bandwidth requirements, which all need further explorations.

ACKNOWLEDGMENTS

This research was supported in part by the Guangdong Natural Science Funds for Distinguished Young Scholars (No. S20120011468), the Guangzhou Pearl River New Star Technology Training Project (No.2012J2200081), Guang-dong National Science Foundation (NSF) Grant (No. S2012010010427), and China NSFC Grant 61202454.

REFERENCES

- [1] K. Tan, J. Fang, Y. Zhang, S. Chen, L. Shi, J. Zhang, and Y. Zhang, "Fine Grained Channel Access in Wireless LAN," Proc. ACM Special Interest Group Data Comm. (SIGCOMM), 2010.
- [2] S. Sen, R. Choudhury, and S. Nelakuditi, "Listen (on the Frequency Domain) Before You Talk," Proc. Ninth ACM Special Interest Group Data Comm. (SIGCOMM) Workshop Hot Topics in Networks (HOTNETS), 2010.
- [3] S. Sen, R. Choudhur, and S. Nelakuditi, "No Time to Countdown: Backing Off in the Frequency Domain," *Proc. ACM MobiCom*, 2011.
- [4] K. Tan, J. Zhang, J. Fang, H. Liu, Y. Ye, S. Wang, Y. Zhang, H. Wu, W. Wang, and G.M. Voelker, "Sora: High Performance Software Radio Using General Purpose Multi-Core Processors," Proc. Sixth USENIX Symp. Networked Systems Design and Implementation (NSDI), 2009.
- [5] S. Katti, S. Gollakota, and D. Katabi, "Embracing Wireless Interference: Analog Network Coding," Proc. ACM Special Interest Group Data Comm. (SIGCOMM), 2007.
- [6] K. Amachandran, E. Belding-Royer, K. Almeroth, and M. Buddhikot, "Interference-Aware Channel Assignment in Multi-Radio Wireless Mesh Networks," *Proc. IEEE INFOCOM*, 2006.
- [7] ANSI/IEEE Std 802.16-2004, IEEE Standard for Local and Metropolitan Area Networks Part 16: Air Interface for Fixed Broadband Wireless Access Systems, ANSI/IEEE, 2004.
- [8] LTE; Evolved Universal Terrestrial Radio Access (E-UTRA); Long Term Evolution (LTE) Physical Layer; General Description, 3GPP Technical Specification TS 36.201-8.2.0, 2009.
- [9] V. Shrivastava, S. Rayanchu, J. Yoon, and S. Banerjee, "802.11n under the Microscope," Proc. ACM Special Interest Group Data Comm. (SIGCOMM), 2008.
- [10] "Gnu Software Defined Radio," Blossom, http://www.gnu.org/ software/gnuradio, 2013.
- [11] Ettus Research, "The Universal Software Radio Peripheral or USRP," http://www.ettus.com, 2008.
- [12] IEEE Std 802.11n-2009, IEEE Standard for Local and Metropolitan Area Networks Part 11; Amendment 5: Enhancements for Higher Throughput, IEEE, 2009.
- [13] Z. Zeng, Y. Gao, K. Tan, and P. Kumar, "CHAIN: Introducing Minimum Controlled Coordination into Random Access MAC," *Proc. IEEE INFOCOM*, 2011.
- [14] S. Gollakota and D. Katabi, "Zigzag Decoding: Combating Hidden Terminals in Wireless Networks," Proc. ACM Special Interest Group Data Comm. (SIGCOMM), 2009.
- [15] K. Wu, H. Tan, Y. Liu, J. Zhang, Q. Zhang, and L. Ni, "Side Channel: Bits over Interference," *Proc. ACM MobiCom*, 2010.
- [16] G. Nychis, T. Hottelier, Z. Yang, S. Seshan, and P. Steenkiste, "Enabling MAC Protocol Implementations on Software-Defined Radios," Proc. Sixth USENIX Symp. Networked Systems Design and Implementation (NSDI), 2009.
- [17] R. Chandra, R. Mahajan, T. Moscibroda, R. Raghavendra, and P. Bahl, "A Case for Adapting Channel Width in Wireless Networks," ACM Special Interest Group Data Comm. (SIGCOMM) Computer Comm. Rev., vol. 38, pp. 135-146, 2008.

Authorized licensed use limited to: SHENZHEN UNIVERSITY. Downloaded on May 11,2023 at 15:14:23 UTC from IEEE Xplore. Restrictions apply.

- [18] D. Halperin, W. Hu, A. Sheth, and D. Wetherall, "Predictable 802.11 Packet Delivery from Wireless Channel Measurements," *Proc. ACM Special Interest Group Data Comm. (SIGCOMM)*, 2010.
- [19] B. Nardelli, J. Lee, K. Lee, Y. Yi, S. Chong, E. Knightly, and M. Chiang, "Experimental Evaluation of Optimal CSMA," *Proc. IEEE INFOCOM*, 2011.
- [20] D.H. Kang, Y. Choi, D. Kim, and S. Choi, "QoS-Aware Load Indicators for Intelligent Cell Selection," Proc. Asia-Pacific Conf. Wearable Computing Systems (APWCS), 2010.
- [21] Y. Bejerano, H. Choi, S. Han, and T. Nandagopal, "Performance Tuning of Infrastructure-Mode Wireless LANs," Proc. Eighth Int'l Symp. Modeling and Optimization in Mobile, Ad Hoc and Wireless Networks (WiOpt), 2010.
- [22] G. Zhou, C. Huang, T. Yan, T. He, J. Stankovic, and T. Abdelzaher, "MMSN: Multi-Frequency Media Access Control for Wireless Sensor Networks," *Proc. IEEE INFOCOM*, 2006.
- [23] J. Zhao, H. Zheng, and G. Yang, "Distributed Coordination in Dynamic Spectrum Allocation Networks," Proc. IEEE First Int'l Symp. New Frontiers in Dynamic Spectrum Access Networks (DySpan), 2005.
- [24] Y. Cheng, H. Li, P. Wan, and X. Wang, "Capacity Region of a Wireless Mesh Backhaul Network over the CSMA/CA MAC," *Proc. IEEE INFOCOM*, 2010.
- [25] J.G. Proakis, Digital Communications, fourth ed. McGraw-Hill, 2001.
- [26] M. Marey and H. Steendam, "Analysis of the Narrowband Interference Effect on OFDM Timing Synchronization," *IEEE Trans. Signal Processing*, vol. 55, no. 9, pp. 4558-4566, Sept. 2007.
- [27] M. Souryal, L. Klein-Berndt, L. Miller, and N. Moayeri, "Link Assessment in an Indoor 802.11 Network," Proc. IEEE Wireless Comm. and Networking Conf. (WCNC), 2006.
- [28] M. Ozdemir and H. Arslan, "Wireless Communications and Networking Conference," IEEE Comm. Surveys and Tutorials, 2007.
- [29] A.J. Viterbi, CDMA: Principles of Spread Spectrum Communication. Addison-Wesley, 1995.
- [30] S. Sen, N. Santhapuri, R. Choudhury, and S. Nelakuditi, "Successive Interference Cancellation: A Back-of-the-Envelope Perspective," Proc. Ninth ACM Special Interest Group Data Comm. (SIGCOMM) Workshop Hot Topics in Networks (HOTNETS), 2010.



Kaishun Wu received the PhD degree in computer science and engineering from the Hong Kong University of Science and Technology (HKUST) in 2011. He is currently a research assistant professor at the Fok Ying Tung Graduate School at HKUST. His research interests include wireless communication, mobile computing, wireless sensor networks, and data center networks. He is a member of the IEEE and the IEEE Computer Society.



Haochao Li received the BEng and MPhil degrees from the Hong Kong University of Science and Technology (HKUST) in 2009 and 2011, respectively. He is currently working toward the PhD degree at HKUST. His research interests include wireless communication, mobile computing, RFID systems, and wireless sensor networks. He is a student member of the IEEE.



Lu Wang received the BEng degree from Nankai University, Tianjin, China, in 2009, and is currently working toward the PhD degree from the Hong Kong University of Science and Technology. Her research interests include wireless communications, including ad hoc networks and cognitive radio networks. She is a student member of the IEEE.



Youwen Yi received the BSc (Hons) degree from the Harbin Institute of Technology in 2007 and the MSc degree from the Huazhong University of Science and Technology in 2009. From 2009 to 2011, he was a research assistant at the Hong Kong University of Science and Technology. Currently, he is a senior research engineer at Huawei Technologies. His research interests include real-time indoor localization systems and next generation self-organized

networks. He is a student member of the IEEE.



Yunhuai Liu received the BE degree in computer science and technology from Tsinghua University, Beijing, in 2000 and the PhD degree in computer science and engineering from the Hong Kong University of Science and Technology (HKUST) in 2008. From 2008 to 2010, he worked at HKUST as a research assistant professor. His research interests include wireless sensor networks, cognitive radio networks, and extreme-scale datacenter and data net-

works. He is a member of the IEEE.



Dihu Chen is a professor in the School of Physics and Engineering and the head of Microelectronic Department, Sun Yat-sen University, China. His research interests include microelectronics and IC design. He is a member of the IEEE.



Xiaonan Luo is a professor in the School of Information Science and Technology and director of the Computer Application Institute, Sun Yat-sen University, China. His research interests include mobile computing, computer graphics, and CAD. He is a member of the IEEE.



Qian Zhang received the PhD degree from Wuhan University, China, in 1999. She joined the Hong Kong University of Science and Technology in September 2005, where she is a full professor in the Department of Computer Science and Engineering. Her current research interests include wireless communications, IP networking, multimedia, P2P overlay, and wireless security. She is a fellow of the IEEE.



Lionel M. Ni is a chair professor in the Department of Computer Science and Engineering at the Hong Kong University of Science and Technology (HKUST). He also serves as a special assistant to the president of HKUST, is dean of the HKUST Fok Ying Tung Graduate School, and is a visiting chair professor of the Shanghai Key Lab of Scalable Computing and Systems at Shanghai Jiao Tong University. He has chaired more than 30 professional confert the IEEE

ences. He is a fellow of the IEEE.