Channel Switching Control Policy for Wireless Mesh Networks

Xiaoguang Li, Jie Wu, Shan Lin, and Xiaojiang Du

Computer and Information Sciences, Temple University, Philadelphia, 19122 USA

Abstract

Dynamic channel assignment algorithms allow wireless nodes to switch channels when their traffic loads exceed certain thresholds. These thresholds represent estimations of their throughput capacities. Unfortunately, the threshold estimation may not be accurate due to co-channel interference (CCI) and adjacent-channel interference (ACI), especially with high traffic loads in dense networks. When the link capacity is over-estimated, these channel assignment algorithms are not effective. This is because channel switch is not triggered even with overloaded data traffic and the link quality decreases significantly as the channel is overloaded. When the link capacity is under-estimated, the link is under utilized. Moreover, when link traffic load increases from time to time, channel switch occurs frequently. Such frequent channel switches increase latency and degrade throughput, and can even cause network wide channel oscillations. In this paper, we propose a novel threshold-based control system, called balanced control system (BCS). The proposed threshold-based control policy consist of deciding, according to the real time traffic load and interference, whether to switch to another channel, which channel should be switched to and how to perform the switch. Our control model is based on a fuzzy logic control. The threshold which assists to make the channel switch decisions, could be deduced dynamically according to the real-time traffic of each node. We also design a novel dynamic channel assignment scheme, which is used for the selection of the new channel. The channel switch scheduler is provided to perform channel-switch processing for sender and receiver over enhanced routing protocols. We implement our system in NS2, and the simulation results show that with our proposed system, the performance improves by 12.3%-72.8% in throughput and reduces 23.2%-52.3% in latency.
1. Introduction

*Wireless mesh networks* (WMNs) are gaining significant momentum as an inexpensive way to provide last-mile broadband internet access. Recent studies [1, 2] have shown that equipping each node with multiple interfaces can improve the capacity of WMNs. By equipping interfaces in different channels, a node can communicate with multiple nodes simultaneously. Each channel allows multiple data flow exchanges in both directions, as long as the traffic load does not exceed the link's throughput capacity, i.e., the maximum amount of traffic that the link can carry.

Many previous research in WMNs usually assume static channel capacity. This simplified assumption does not hold in reality. The throughput capacities in real systems can vary dramatically with time and locations due to fading, shadowing, and interference. As a result, protocols based on static channel capacity may not work well in real systems as channel throughput capacity (or simply link capacity) can be either over-estimated or under-estimated.

Static channel assignment algorithms that switch channels periodically or permanently [3, 4], have been shown to achieve great performance with stable network traffic. However, with dynamic traffic loads, such algorithms are not effective due to the mismatch between dynamic channel throughput capacity and the real-time traffic load. To select a channel based on real-time traffic load, recent studies on dynamic channel assignment algorithms [5, 6, 7] can adaptively switch the channel on certain links in a distributed fashion. Accurate estimation of channel throughput capacity is very challenging, as it is notably influenced by both co-channel interference (CCI) [8] and adjacent-channel interference (ACI) [9], especially when the traffic load is high. When the link capacity is over-estimated, the channel saturates and the channel quality degrades before channel switch is triggered. On the other hand, when the link capacity is under-estimated, channel is not fully utilized. Also, when the traffic load experiences temporary increases, existing algorithms tend to switch the channel frequently. Such frequent channel switches degrade the network throughput and increase latency significantly. Moreover, the newly switched links cause interference to other nearby links, introducing link
capacity variation on those links and triggering even more channel switches. In the worst case, it can cause network wide channel oscillation.

An intuitive example is shown in Figure 1. With an over-estimated threshold, the channel switch is not triggered in all cases, even when the channel saturates and the link quality degrades. While with an under-estimated threshold, the channel capacity is not fully utilized and channel switches occur frequently (at time 2, 4, 6, 7, and 8). If we can choose the link capacity threshold adaptively, the channel is better utilized than using the under-estimated threshold in all the cases. Moreover, the overhead in channel switching is reduced. We note that existing rate adaptation protocols[10, 11] adjust transmission rate based on channel contention. Our work focuses on channel switching, which is an orthogonal issue to rate adaptation.

Our goal is to dynamically find a channel capacity estimation that fully utilizes link capacity and reduces unnecessary channel switching. In this paper, we propose a threshold based control system, called balanced control system (BCS). Unlike existing approaches that use static threshold estimation, our design features a fuzzy control loop to monitor the dynamic traffic load during runtime and adaptively adjust the channel switching threshold. This threshold serves as the bound for traffic load on this channel. The proposed threshold-based control solution consists of deciding, according to the real time traffic load and interference, whether to switch to another channel, which channel should be switched to and how to perform the switch. Our
threshold control model is based on a fuzzy logic control. The threshold which assists to make the channel switch decisions could be deduced dynamically according to the real-time traffic of each node. Our control based design allows the dynamic threshold to approximate the runtime capacity accurately, therefore improving the channel utilization and reducing unnecessary channel switches. The contributions of our work are demonstrated as follows:

1. We propose a threshold-based balanced control system (BCS) in which each link in the network finds its own threshold according to the real-time traffic. We also offer a traffic metric model for our BCS. The metric model estimates the traffic load integrated with CCI and ACI problems.

2. We present a dynamic channel assignment scheme for the selection of the new channel. This algorithm fully utilizes variable channel capacities with reduced channel switching overhead. We also provide a channel switch scheduler to perform channel-switch processing for sender and receiver over enhanced routing protocols.

3. We implement our system in NS2. From our simulation results, we demonstrate that our proposed scheme outperforms the current techniques. Although channel switch algorithms for wired networks have been studied and practiced in industry [12]. In wired networks, static channel capacity models are highly accurate. Whereas in wireless networks, fading and interference (ACI and CCI) can cause significant channel throughput variations, resulting in frequent channel switch.

The remainder of this paper is organized as follows: Section 2 describes our proposed balanced control system. Section 3 provides evaluation results. Section 4 gives related work. The paper concludes in Section 5.

2. Balanced control system

In this section, we present the problem formulation, problem statement, and our system model. Our system is composed of traffic metric model, fuzzy control model, dynamic channel switch scheme, and channel switch scheduler.

2.1. Problem Formulation

In this subsection, we describe the model formulation for our system in WMNs. Recall that WMNs consist of a set of stationary wireless routers,
some of them acting as gateways to the Internet. Specifically, we do not require the presence of special gateway nodes, which could be the source or destination of all traffic in the network.

We define our system requirement as follows: (1) Two nodes that can communicate with each other should have at least one common channel. (2) We assume that every node in our system uses a single channel to communicate with each neighbor. (3) The common default channel is required for transferring control messages and is used as a temporary channel for data transfer. (4) Channels refer to different frequency bands. All of the channels are working on the half duplex mode.

2.2. Problem Statement

Channel assignment algorithms in WMNs select channels for each link in the network in order to optimize network throughput and reduce latency. Recent channel assignment research explores node’s ability of dynamically switching channels. In essence, when the total amount of traffic load along this link exceeds the link capacity, nodes can either reduce the traffic load on this link or switch the channel. If the channel capacity is degraded due to CCI and ACI, it is desirable to switch from the current channel to another channel with higher bandwidth. However, channel switching incurs noticeable latency due to synchronization overhead between a pair of nodes, which also decreases the link throughput. Therefore, there is a tradeoff between the benefit of channel switching and its overhead.

In this paper, we explore when to perform channel switching under dynamic channel throughput. Previous research on channel switch is usually based on analysis with static channel models. Although these works provide valuable insights on this problem, unfortunately, these assumptions may not be hold in real WMNs: First, wireless link capacity is sensitive to distance and surrounding environment. In WMNs with fixed topology, even though the distance between any pair of nodes is fixed, the link capacity may vary due to environmental changes. Second, the traffic loads sometimes experience transient increases, which will result in the decrease of the traffic load and interference. These problems are not well modeled in previous studies. Third, the traffic loads affect link capacity, especially in dense networks with high traffic load. When traffic load of a link increases, it can cause channel capacity degrade on itself and other nearby links, even they are assigned with different channels due to interference. Therefore, with the dynamic traffic loads and channel capacity, previous solutions may not work well.
2.3. System Overview

Our solution to the above problem has four key components: a traffic metric model, a fuzzy control model, a dynamic channel assignment algorithm, and an enhanced routing algorithm. In the fuzzy control model as shown in Figure 2, a control loop is designed to monitor the channel capacity and dynamically adjust the threshold. The threshold serves as a condition for channel switching algorithm. Our control model provides a reasonable fuzzy control mechanism for deciding whether channel switching should be performed and which channel should be switched to.

Based on the proposed model, the dynamic channel assignment algorithm performs channel switch during runtime in a distributed fashion. The conditions for selecting new channels are specified in the fuzzy control model. With the support of channel switch control and channel assignment, enhanced routing algorithms can achieve better performance than previous solutions.

2.4. Traffic Metric Model

There are significant amount works focusing on the model of interference level [13, 14]. In this subsection, we offer an integrated model to estimate the level of traffic load and interference on each link.

Assuming $c$ is the current channel, and $i$ is the current node, if $j$ is the neighbor of node $i$, then, the traffic load between nodes $i$ and $j$ is the sum of all the outing flow $flow_t$ and incoming flow $flow_r$, as described by Equation (1):

$$bw^c(i) = \sum_{t=1}^{F_t} flow_t + \sum_{r=1}^{F_r} flow_r \quad (1)$$

To solve the CCI problem, we use the following Equation (2) to describe
Table 1: Traffic metric model table

<table>
<thead>
<tr>
<th>Notation</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>pair$(i, j)$</td>
<td>a link between node $i$ and its neighbor $j$</td>
</tr>
<tr>
<td>$bw^c(i)$</td>
<td>traffic load of node $i$ on channel $c$</td>
</tr>
<tr>
<td>$F_t$</td>
<td>number of flows going through $pair(i, j)$</td>
</tr>
<tr>
<td>$t, r$</td>
<td>flow id for transmitting/receiving the packets</td>
</tr>
<tr>
<td>$flow_t, flow_r$</td>
<td>traffic load of outgoing flows $t$ or incoming flows $r$</td>
</tr>
<tr>
<td>$n_i$</td>
<td>neighbor id for node $i$</td>
</tr>
<tr>
<td>$bwt^c_{i,j}$</td>
<td>total traffic load of $pair(i, j)$ on channel $c$</td>
</tr>
<tr>
<td>$N_i$</td>
<td>number of neighbors of node $i$</td>
</tr>
<tr>
<td>$\gamma$</td>
<td>interference ratio for ACI</td>
</tr>
<tr>
<td>$c, c_n$</td>
<td>node’s current channel and adjacent channel</td>
</tr>
<tr>
<td>$btt^c_{i,j}$</td>
<td>traffic metric of channel $c$ which integrates CCI and ACI</td>
</tr>
</tbody>
</table>

the total traffic load on channel $c$ in two hops for the $pair(i, j)$:

$$bwt^c_{i,j} = \sum_{n_i=1}^{N_i} bw^c(n_i) + \sum_{n_j=1, n_j \neq i}^{N_j} bw^c(n_j),$$

(2)

where $N_i$ and $N_j$ are the number of neighbors of node $i$ and node $j$ in the network, respectively. This equation is used to derive the total traffic load of $pair(i, j)$ and their neighbors, where they work on the current channel $c$.

The usage of adjacent channels can cause interferences [9]. Then, we can obtain the interference ratio $\gamma$ according to the current channel measurement for ACI:

$$\gamma = 1 - \frac{1}{c_n} \frac{|c - c_n|}{c_n},$$

(3)

where $c$ is the current channel, which is used by node $i$ and $c_n$ is the adjacent channel corresponding to other interfaces of node $i$. The overall metric
can be deduced as follows which integrates traffic load and adjacent channel interference factors based on two hops neighborhood information:

\[ \text{btt}_{i,j}^c = \text{bwt}_{i,j}^c + \sum_{l=1}^{N_i} (\gamma \times \text{bwt}_{n_i;l}^c) \]  

(4)

Thus, we have offered our traffic metric model. Note that our traffic model is based on the estimation of traffic load and interference level.

2.5. Fuzzy Control Model

In this subsection, we present our system model. Our control model provides a reasonable fuzzy control mechanism for deciding which channel to be switched to. The goal is to find a per-link threshold for channel switching. An ideal threshold will approximate run-time channel throughput and reduces the unnecessary channel switchings. In our fuzzy control model, if the threshold is set too small which indicates frequent switches, we will increase the threshold value. In contrast, if the threshold is set too large which prevents the system from performing beneficial channel switch; we will lower the threshold value accordingly. Therefore, we need to find the threshold adapting to the link capacity changes.

In this paper, we use fuzzy control to perform such an adjustment because it is difficult to accurately estimate the right threshold value a priori. Fuzzy control offers a convenient method for constructing nonlinear controllers via the use of heuristic information [15]. We present our fuzzy control model to adjust the threshold value. Accurate threshold adjustment is important because small errors in the thresholds can induce large channel-switch overheads. However, it is very challenging to directly estimate the exact threshold value since the environment is constantly changing. Thus, we use a fuzzy control model to demonstrate our strategy, which is similar to the automobile "cruise control" example in [16]. In our design, fuzzy interpretations are extended using the fuzzy set theoretic operations [17].

In our fuzzy control model, we use the interval time \( \tau \) as a timer to monitor the network state. If the interval time between current switch and previous switch happens within the interval time \( \tau \), we will increase the threshold value after this switch. Otherwise, we will decrease the threshold value. We use Equation (5) to express this idea. There are two parts in this equation: the first part is that the current interval time is less than interval time \( \tau \). When the current bandwidth is larger than constraint bandwidth
Table 2: Fuzzy control model table

<table>
<thead>
<tr>
<th>Notation</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>( m )</td>
<td>switching times</td>
</tr>
<tr>
<td>( \mu_{i,j}(m) )</td>
<td>threshold for pair ((i, j)) with number of switching times ( m )</td>
</tr>
<tr>
<td>( \Delta t_{i,j}(m) )</td>
<td>interval time between current and previous switches</td>
</tr>
<tr>
<td>( \phi_{i,j}(m) )</td>
<td>control weight of threshold with switching times ( m )</td>
</tr>
<tr>
<td>( t_{i,j}(m) )</td>
<td>the time when the pair ((i, j)) switches to a new channel</td>
</tr>
<tr>
<td>( E(m) )</td>
<td>the value for deviation with switching times ( m )</td>
</tr>
<tr>
<td>( R(m) )</td>
<td>the variance ratio with switching times ( m )</td>
</tr>
<tr>
<td>channelList</td>
<td>available channels in the network</td>
</tr>
<tr>
<td>channel((i)(j))</td>
<td>the channel on the interface ( j ) of node ( i )</td>
</tr>
</tbody>
</table>

\( \mu_{i,j}(m) \) where \( m \) is the number of switches, we need to switch the channel, and also increase the \( \mu_{i,j}(m) \); the second part is that the current time is larger than interval time \( \tau \). This means there is no switching during this interval. We need to lower the threshold value:

\[
\begin{cases}
\mu_{i,j}(m) = \mu_{i,j}(m - 1) + \phi_{i,j}(m - 1), & \Delta t_{i,j}(m) \leq \tau \\
\mu_{i,j}(m) = \mu_{i,j}(m - 1) - \phi_{i,j}(m - 1), & \text{otherwise} \\
\mu_{i,j}(0) = \mu_0, \phi_{i,j}(0) = 0,
\end{cases}
\]

where \( \phi_{i,j}(m) \) is the control weight and \( \mu_{i,j}(x) \) is the constraint bandwidth for pair \((i, j)\). \( \mu_{i,j}(0) \) and \( \phi_{i,j}(0) \) are the initial values of the two parameters. The threshold can be improved during a interval time \( \tau \). As we can see from this equation, the control weight is important for the controlling process. That is how much we need to increase or decrease the threshold value. Moreover, when the switch should be performed is also an important issue. For this we use the Equation 6 to demonstrate the control weight \( \phi_{i,j}(m) \):

\[
\phi_{i,j}(m) = \alpha_{i,j}(m) \times E(m) + (1 - \alpha_{i,j}(m)) \times R(m - 1)
\]
where $E(m)$ is the value for deviation, and $R(m)$ is the variance ratio. $\alpha_{i,j}(m)$ is the weight for the balanced formula. $E(m)$ and $R(m)$ have different effects among different switches. Note that $E(m)$ should be much larger than $R(m)$. We use $\alpha_{i,j}(m)$ to adjust the threshold. Sometimes, we want to adjust this threshold quickly, thus, we increase the $\alpha_{i,j}(m)$ for $E(m)$. Other times, we prefer that it changes slowly, thus, we decrease the $\alpha_{i,j}(m)$ for $R(m)$. Then, we can use the weight $\alpha_{i,j}(m)$ to control the two parts. Next, we offer the following equation to obtain $E(m)$, $R(m)$ and the weight $\alpha_{i,j}(m)$.

\[
\begin{align*}
E(m) &= \mu_{i,j}(m) \\
R(m) &= \frac{\phi_{i,j}(m)}{(t_{i,j}(m) - t_{i,j}(m-1))} \\
\Delta t_{i,j}(m) &= t_{i,j}(m) - t_{i,j}(m-1) \\
t_{i,j}(0) &= 0, \forall i, j, m \in N, \tau > 1, \\
\alpha_{i,j}(m+1) &= |1 - \frac{t_{i,j}(m) - t_{i,j}(m-1)}{\tau}| \Delta t_{i,j}(m) \leq \tau \\
\alpha_{i,j}(m+1) &= 0, \text{otherwise}
\end{align*}
\]

(7)

where $t_{i,j}(m)$ is the recorded time when the pair $(i,j)$ switches to a new channel. For the weight $\alpha_{i,j}(m+1)$, if the interval time between the current switch and previous switch is larger, the current threshold is close to the estimated threshold. Then, the weight of $R(m)$ is larger according to the weight $\alpha_{i,j}(m+1)$. Otherwise, the weight of $E(m)$ is larger.

Below, we summarize the whole process. At the start of the controlling process, the deviation $E(m)$ is more important. During this period, the control weight $\phi_{i,j}(m)$ can help the system quickly find the range of the threshold. A decrease in the $\alpha_{i,j}(m)$ will make the variance ratio $R(m)$ become the main factor of the system and $E(m)$ become less important. Thus, it could adjust the value, and control the estimated threshold in this range.

2.5.1. Stability Analysis

**Theorem 1.** If the control weight can be infinitely close to 0, then our system could finally find the estimated threshold value $\mu_{i,j}(m)$. Thus, we need to prove the following conclusion:

\[
\lim_{\Delta t \to \tau, m \to \infty} \phi_{i,j}(m) = 0
\]
Proof. From Equation (6), the problem can be converted to two parts:

\[
\lim_{\Delta t \rightarrow \infty, m \rightarrow \infty} \alpha_{i,j}(m+1) \times E(m) = 0,
\]
\[
\lim_{\Delta t \rightarrow \infty, m \rightarrow \infty} (1 - \alpha_{i,j}(m+1)) \times R(m) = 0
\]

Here, we know:

\[E(m) < C,\text{ such that } C \in \mathbb{R}\]

Although \(C\) could be sufficiently large, \(C\) actually is a finite number. Thus, for the first part, we only need to prove:

\[
\lim_{\Delta t_{i,j} \rightarrow \tau, m \rightarrow \infty} \alpha_{i,j}(m+1) = 0
\]

From Equation 7, we have:

\[
\lim_{\Delta t_{i,j} \rightarrow \tau, m \rightarrow \infty} \alpha_{i,j}(m+1) = \lim_{\Delta t \rightarrow \tau} |1 - \frac{\Delta t(m)}{\max(\Delta t_{i,j}(m), \tau)}| = \lim_{\Delta t_{i,j} \rightarrow \tau, m \rightarrow \infty} \phi_{i,j}(m) \Delta t_{i,j}.
\]

According to the first part of Equation (5), we know that the interval time \(\Delta t_{i,j}(m)\) is increased after the switch, since the threshold is larger and more difficult to meet switching conditions. However, in the second part, the proof is obvious, since \(\frac{\Delta t_{i,j}(m)}{\max(\Delta t_{i,j}(m), \tau)} = 1\). Thus, we prove that

\[
\lim_{\Delta t_{i,j} \rightarrow \tau, m \rightarrow \infty} \alpha_{i,j}(m+1) = 0
\]

Next, we need to verify the following assumption:

\[
\lim_{\Delta t \rightarrow \tau, m \rightarrow \infty} (1 - \alpha_{i,j}(m+1)) \times R(m) = 0
\]

Because we have:

\[
\lim_{\Delta t \rightarrow \tau, m \rightarrow \infty} (1 - \alpha_{i,j}(m+1)) \times R(m) \leq \lim_{\Delta t \rightarrow \tau, m \rightarrow \infty} R(m)
\]
\[
= \lim_{\Delta t \rightarrow \tau, m \rightarrow \infty} \frac{\phi_{i,j}(m)}{\Delta t_{i,j}}.
\]

\(\phi_{i,j}(m)\) could be sufficiently large. It is actually bounded by a finite value \(\phi_{\text{max}}\), such that \(\phi_{i,j}(m) < \phi_{\text{max}}\). According to Cauchy series [18], monotone
sequence converges if and only if it is bounded. Since $\Delta t_{i,j}$ is increased during the controlling process, then:

$$\lim_{\Delta t \to \tau, m \to \infty} \frac{\phi_{i,j}(m)}{\Delta t_{i,j}} = 0$$

Thus, this concludes the proof of our solution. \qed

2.5.2. Examples

In this part, we will offer an example of our proposed model. We will formalize the theory results according to our analysis. We first change Equation (6) to another form. We set:

$$A = \alpha_{i,j}(m + 1) \times \mu_{i,j}(m)$$

$$B = \frac{1 - \alpha_{i,j}(m + 1)}{t_{i,j}(m) - t_{i,j}(m - 1)}$$

Then, Equation (6) can be deduced:

$$\phi_{i,j}(m + 1) = A \cdot (\mu_{i,j}(0) + \sum_{k=0}^{m-1} \phi_{i,j}(k)) + (A + B) \cdot \phi_{i,j}(m) \quad (9)$$

Figure 3 demonstrates theoretical results of formula $\phi_{i,j}(m)$ (see Equation (9)). Because other values could be deduced after each switch, we only need
to consider $A$ and $B$ of Equation (9), because $\Delta t_{i,j}(m)$ of $B$ is the unknown result measured from the real-time record. We set the value of $\Delta t_{i,j}(m)$ from 1 to 20. The increase is 1 each time. These values could be different each time, but they must be incremental. The reason is that each time the threshold increases, it becomes more difficult to meet the channel switch condition. Without the loss of generality, we set $\tau = 20$ and $\mu_{i,j}(0) = 100$. From Figure 3, we can see that $\phi_{i,j}(x)$ first increases and then decreases as time goes on. This confirms the validity of our design. The purpose of the control system is to dynamically find the threshold and make the system stable. To achieve this goal, we should continuously reduce the selected area of the constraint $\mu_{i,j}(m)$ until we find the exact value $\mu_{i,j}(m)$. In other words, we first reduce the area as quickly as possible, and then adjust it. The increased process of $\phi_{i,j}(x)$ is to find the smallest area. In addition, the decreased process is to adjust the value in the small selected area. That is the reason we need the control weight $\phi_{i,j}(x)$ to be first increasing and then decreasing.

2.6. A Dynamic Channel Assignment Scheme

We design a dynamic channel assignment algorithm (GNOC) to get the next optional channel for our system. The GNOC is used to select new channels for transmission. We will select the channel which is lower than the threshold and with minimum ACI. Table 3 offers the notation for the GNOC algorithm.

In Algorithm 1, the channelList is the available channels in WMNs. The current channel, say $c$, is the current channel used for transmission. We first construct a temporary channel list, say tempList, from the available channel
Algorithm 1 GNOC

1: let $\text{tempList} = \text{cList}(i)$;
2: for $(k = 0; k < N_i; k++)$ do
3:     if $\text{channel}(j)(k)$ is not in the $\text{tempList}$ then
4:         $\text{channel}(j)(k)$ is added into $\text{tempList}$
5:     end if
6: end for
7: for $(j = 0; j < \text{size of channelList}; j++)$ do
8:     for $(k = 0; k < \text{size of tempList}; k++)$ do
9:         $c$: get $k$th channel from $\text{tempList}$
10:        $\text{diff} = \frac{\text{size of channelList}}{N_i}$;
11:        $\text{diff} = \text{min}(\text{diff, abs(channel}(i)(j) - c)))$
12:     end for
13: end for

list of node $i$. Then, we get node $i$’s neighbor list. To each neighbor of node $i$, we add the channels that the neighbors of node $i$ are using to the $\text{tempList}$. Then, we obtain the absolute value (abs) according to the channel in the $\text{channelList}$ and $\text{tempList}$. We set $\text{diff}$ to be an extreme value of differences among channels. Then, the final channel is the channel with a minimum $\text{diff}$.

The new channel we get is the channel with least ACI. However, we are not sure about the traffic statement of the new channel. Therefore, we also check the current statement of the new channel according to our proposed routing metric. If it does not exceed the threshold $\mu_{i,j}(x)$, then the new channel is the next optional channel. Otherwise, we remove this channel and run Algorithm 1 again until we find it.

2.7. Enhanced Routing Protocols in WMNs

First, we demonstrate why the existing routing protocol can not be used in our scheme. The existing routing protocols, such as AODV [19] and FSR [20] support multiple interfaces. However, these designs are typically used for the multi-home based protocol (such as SCTP and DCCP) instead of used for multiple interface wireless mesh networks.

We propose to introduce another field called “channel id” to the routing table entry. This “channel id” can be used among neighboring nodes
to coordinate their channel selection processes. The two neighbors along a channel can talk to each other, when they switch to the same channel, specified by “channel id”. When the node has switched to another channel, and the “channel id” has changed, its routing table should be updated according to the new “channel id” information.

In our model, we propose modifications to the current routing protocols, AODV and FSR, so as to enable the discovery of channel information from a source to a destination. The proposed enhancement can help the node find the correct route, even if the nodes have switched channels.

E-AODV routing: When the source node requests to send the packet to the destination, it will send “hello” information. The broadcast packet RREQ will be flooded to every interface of the node. Channel information is also added into the packets, and the corresponding nodes that receive the notice will update the routing table entry. At that time, the “channel id” is updated, and the nodes will select the correct route to communicate.

E-Fisheye routing: In the FSR scheme, we update the routing table according to the channel table. If the channel table has been changed, the routing table needs to be updated as well.

2.8. Channel Switch Scheduler

The channel and interface information has been maintained by every node in WMNs. TABLE 3 details the information for every node.

The $\text{neighbor}_\text{Set}(i)$ is the set for all the neighbors of the current node, say node $i$. The number of available channels contains all the channels that can be used for the whole network. The channel group here is the $\text{channel}_\text{Group}(i)(l)$, which is the channel for the interface $l$ of node $i$. Every node in this network can switch channels when it satisfies the threshold $\mu_{i,j}(m)$. Moreover, special conditions need be satisfied:

- **Condition 1**: There is another common channel for the $\text{pair}(i,j)$. If the traffic load of that channel does not exceed threshold $\mu_{i,j}(m)$, then the $\text{pair}(i,j)$ will select that channel to communicate.

- **Condition 2**: If the new channel, taken from channel assignment algorithm 1, is already working for another interface of node $j$, then only node $i$ switches to the new channel.
Algorithm 2 Executed by sender

1: if $btt_{i,j}^c < \mu_{i,j}(m)$ then
2:   forward packet;
3: else
4:   if the number of common channels between node $i$ and node $j$ is larger than 1 then
5:     update the current routing table entry
6:   go to step 1
7: end if
8: if unused interfaces of node $i$ is larger than 0 then
9:   newchannel = interface($i$)
10: go to step 1
11: end if
12: newchannel = the next optional channel according to algorithm 1 (GNOC);
13: if $btt_{i,j}^c < btt_{i,j}^{newchannel}$ then
14:   newchannel = $c$
15: end if
16: step 1:
17: update $\mu_{i,j}(m)$ according to Equation (5);
18: neighborNodeList = $yList(x)$
19: generate chr.packet
20: update the current routing table entry
21: send chr.packet to the nodes in $ylist(x)$;
22: end step 1;
23: forward packet using the new routing table
24: end if

- Condition 3: There are other nodes that are neighbors of $pair(i, j)$. All of them work on the same old channel. These nodes construct a temporary list called $yList$.

If a new pair is added to the transmission, the traffic load exceeds the threshold value of $\mu_{i,j}(m)$ from Equation (5). For example, the $pair(i, j)$ in channel $c$ first checks the condition 1. If it is not satisfied, it will temporarily take channel $c$ from Algorithm 1. Then, it will check condition 2 and get the node to switch. If there is an available interface for the node, it will select the available interface.
Algorithm 3 Executed by receiver

1: receive a chr_packet;
2: if (nodeId ∈ neighborodelist) then
3:    switch channel and update $\mu_{i,j}(m)$ according to Equation (5);
4:    update the current routing table entry;
5:    update channel table;
6: else
7:    update channel table;
8: end if
9: free chr_packet;

Table 4: CHR Packet for Channel Switch Scheduler

<table>
<thead>
<tr>
<th>Notation</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>nodeId</td>
<td>The current node to be switched</td>
</tr>
<tr>
<td>othernodeId</td>
<td>The other node of the pair to be switched</td>
</tr>
<tr>
<td>oldchannelId</td>
<td>The old channel of the pair</td>
</tr>
<tr>
<td>newchannelId</td>
<td>The new channel to be switched for the pair</td>
</tr>
<tr>
<td>neighborodelist</td>
<td>The neighbors sharing the same channel</td>
</tr>
</tbody>
</table>

Since we use the enhanced routing protocols, the channel information can be obtained from the routing table. In unicast, the current node will collect the channel state information through the channel information before transmission. The current traffic load can be obtained from Equation (4). Then, we apply the channel switch algorithm, shown in Algorithm 2 (sender) and 3 (receiver). If the current traffic load exceeds the threshold $\mu_{i,j}(x)$, it will switch to another channel. Before that, the node should send the message to every node in the network. A field `chr_packet` (see Table 4) is added to a packet to carry the information. If the other node in the `neighborodelist` receives the `chr_packet`, and it does not satisfy condition 2, then it is required to switch to the new channel $c$.

We first apply an initial random channel assignment according to the topology generator. The GNOC strategy is used to select the next optimal channel for this switch. The enhanced routing agent is used for the channel
switch scheduler and the proposed balanced controlling model is used to make the decision of the switch.

3. Performance Evaluation

In this section, we evaluate the performance of our proposed system through simulation. We implemented our solution: channel switch control and dynamic channel assignment algorithm in NS-2 simulator. As NS-2 provides rich physical layer models, the NS-2 based simulation have been widely used in research studies. We test the enhanced routing algorithms: AODV and FSR on top of our solutions, which are default routing protocols according to 802.11s [21]. AODV [19] is a reactive routing protocol while FSR [20] is a proactive routing protocol. FSR controls its update overhead using a policy of non-uniform frequency for update. The inner scope nodes are updated more frequently (and hence have more accurate information) than the outer scope nodes. Our solutions can also work with other routing protocols in WMNs.

In the simulation of WMNs, there are several available extensions [4] for M2WMNs. We extend the existing work with switching abilities using NS2. Extensively simulation results demonstrate that our algorithm outperforms existing solutions with static threshold. Overall, our solution improves existing solutions by 12.3%-72.8% in throughput and reduces 23.2%-52.3% in latency.

3.1. Simulation Setup

We select the two-ray ground reflection model. The transmission range is 22 meters, so two nodes that are 22 meters apart can communicate with each other. The listening range is 44 meters, so nodes that are within 44 meters can cause interference to each other. We adopt KN-CA in our evaluation. There are 12 channels in the 802.11b network.

We evaluate our BCS with two enhanced routing protocols: AODV and FSR. With the 802.11b environment, the actual maximum throughput $B_{th}$, with MSDU size of 200 bytes, is 1.21 Mbps. The range of $\mu_{i,j}(x)$ is from 0.49 Mbps to 3.848 Mbps according to interfaces per node. We select 491,510, 891,510, and 1,291,510 bytesps (bytes per second) as the initial values. This traffic profile is fixed for all the simulations. There are 25 nodes in this simulation, and each node has up to five interfaces. Four of these interfaces can be switched for data transmission and one interface is fixed as the default.
control channel. Besides the default control channel, we test two interfaces (2-nics) and three interfaces (3-nics) for data transmission in Section 3.2. In the rest of evaluations, we test four interfaces (4-nics) for data transmission.

3.2. Performance Evaluation of Routing Protocols

We consider different number of interfaces as our evaluation study. Due the space limitation, we focus on comparison between E-AODV and AODV. The comparison result between E-FSR and FSR has the same trend. We use an 802.11b network environment. We adopt an interference based static channel assignment (KN-CA) and AODV routing in a simulation study. The evaluation consists of the three interfaces (3-nics) and 2 interfaces (2-nics). Figure 4 gives the comparison results of the proposed AODV routing and enhanced AODV routing (E-AODV).

As Figure 4 shows, the throughput of our enhanced AODV routing is higher than the original AODV. The 3-nics AODV is better than the 2-nics AODV. The reason is because with enhanced AODV, the selected shortest route is calculated by the common channel. All the pairs of the route are working in the same channel. With 30 heavy traffics, the 3-nics AODV can partake in three different interfaces. The throughput can also be improved.

We also can see that our E-AODV achieves higher reliability than AODV, both in 2-nics and 3-nics cases in regards to the packet loss rate comparison (see Figure 5).

From the above simulation results, we can see that our enhanced scheme for routing protocols performs better in multiple-channel multiple-interface
environments. Then, we evaluate our balanced controlling system in the following sections. We will evaluate the BCS with both E-AODV and E-FSR routing protocols, separately.

3.3. Performance Evaluation with Enhanced-AODV

In our solution, if the traffic load is high, the constraint $\mu_{i,j}(x)$ should be updated for that link. This link also needs a new channel that has less interference.

However, it is not easy to determine the exact value $\mu_{i,j}(x)$. So, BCS will try to get $\mu_{i,j}(x)$ as quickly as possible, according to Equation 5. The simulation time is 80 seconds, and 30 heavy traffics are added separately, with the interval 0.4 seconds. Figure 6 shows the comparison results of the BCS, and without the channel switch control (NCS). The value behind the name in the figures is the initial value of $\mu_{i,j}(0)$.

It is obvious from the throughput comparison that with the same traffic profile, when the system uses the balanced control strategy, the network performance is stable, and also better than the system without the BCS by 40%-70%. With different initial $\mu_{i,j}(0)$, the value of 491,510 bytesps is better than 891,510 bytesps. The reason is that the smaller $\mu_{i,j}(x)$ causes more switches, and the larger value $\mu_{i,j}(x)$ is difficult to achieve. This smaller $\mu_{i,j}(0)$ causes the default channel to take over some of the traffics. This is desirable because these invalid channel switches decrease the network performance. With a BCS and the parameter 491,510 bytesps, the system achieves the best performance using the same traffic profile.

Figure 5: Packet loss rate comparison between AODV and E-AODV
3.4. Performance Evaluation with Enhanced-FSR

As Figure 7 demonstrates, our control system is also better when compared to NCS after 15 seconds. The performance of $\mu_{i,j}(0)$ is better than $\mu_{i,j}(0) = 891,510$. The reason is that 491,510 bytesps is the value that can be easily achieved, thus, some of the traffic is divided to the default channel.

Figure 8 gives the comparison results of the packet delay. We can see that the packet delay increased before 15 seconds. The performance is similar among the four situations. This is because the traffics are added as time goes on. Therefore, the packet delay increases. We can also see that when all the traffics are stable after 15 seconds, the BCS will balance the traffics. It is clear that after 25 seconds, the performance is better when it is used with the BCS.
Also, the system of $\mu_{ij}(0) = 491, 510$ is outperforming $\mu_{ij}(0) = 891, 510$, both with the BCS and without the BCS.

### 3.5. Performance Comparison with Different Enhanced Routing Protocols

This part will demonstrate the comparison between enhanced AODV and FSR routing protocols with the parameter $\mu_{ij}(0) = 1,291,510$. We have shown the results of $\mu_{ij}(0) = 491, 510$ and $891, 510$. We can roughly see the throughput results above in Figure 6 and Figure 7. The performance of the FSR protocol is better. To further verify the result, we select another parameter $\mu_{ij}(0) = 1,291,510$ and compare its performance to the one without channel switch control.
Figure 9 shows the throughput comparison with different routing protocols: AODV and FSR. The performance of the FSR routing protocol is better than AODV routing protocol.

We also present the packet delay and packet loss rate comparison in Figure 10. Since our solution continuously adjusts the channel according to the bandwidth and interference, the packet delay has also been decreased. The packet delay is increased before 15 seconds, and decreased thereafter. The reason for the situation is that we add the traffic, flow by flow, with 0.4 second intervals. After 15 seconds, 30 flows are stable in the network system, and no more traffic will be added in. But, our solution still adjusts the traffic until no more bandwidth exceeds $i_{ij}(x)$. With the same condition, the channel switch control can quickly find the right parameter, making it more efficient. Also, the packet delay is lower with the FSR routing protocol. The simulation results are summarized as follows:

1. Both E-AODV and E-FSR have superior performance than regular AODV and FSR, respectively.
2. Our proposed balanced control system makes the system more efficient than the normal system in a dynamic environment. The simulation results show that throughput, packet loss rate and packet delay are all better than the system without control.
3. The initial parameter is usually difficult to decide. However, in our simulation study, the system performs better when working with a lower parameter $\mu_{i,j}(0)$.
4. Considering different routing protocols working with our channel switch control, the FSR routing protocol performs better than the AODV routing protocol.

4. Related work

Extensive studies have been done to utilize multiple channels in WMNs. Some works focus on changing MAC protocols [22, 23]. In [22], a busy tone is used to show the channel reserving information. However, this MAC protocol cannot be applied directly because it is not compatible with commodity hardware. Protocols of [23] seek to use one interface to exploit multiple channels to improve network performance.

So and Vaidya et al. [24] propose an architecture for multi-channel networks that uses a single interface. Each node has a default channel for receiving data. A node with a packet to transmit has to switch to the channel of the receiver before transmitting data. However, the proposal does not consider the effect of channel switching. The packet has to wait for the delay of the transmission.

To assign channels to the interfaces, [3] presents a localized greedy heuristic based on the interference cost function defined for pairs of channels. [25, 26] consider WMNs with main traffic flowing to and from a gateway, which is also in charge of the channel computation. In their channel assignment to a non-default radio, nodes closer to the gateway and/or bearing higher traffic load receive a better quality channel.
Raniwala et al. [4, 27] devise routing and interface assignment algorithms for mesh networks. The protocols are designed to be used in WMNs, where traffic is directed toward specific gateway nodes. Raniwala's protocols assume traffic load between all nodes is already known unlike our work. Moreover, with the load information, interface assignments and route computations are intelligently computed.

When these models are applied in real systems, the impacts of dynamic environment and interference can cause link capacity estimation to be inaccurate, result in unstable performance.

Wu et al. [28] describe the design, implementation and evaluation of WMN system. That system supports both dynamic channel switching and load-balancing/fault-tolerant routing, and successfully runs on low-cost commodity IEEE 802.11-based access points.

However, few of the studies focus on the switching strategy and dynamic threshold to control the switching process in WMNs. In this paper, our attempt is to study fuzzy control and integrate this model into our system to dynamically find the threshold according to the real-time traffic. The aim of our system is to incorporate expert human knowledge in the control algorithm. In this sense, a fuzzy controller may be viewed as a real-time expert system to balance the unstable network. The only work related with fuzzy logic is discussed in [29]. However, that work is based on the QoS considerations for multimedia transmission.

5. Conclusion and Future Work

In this paper, we proposed a novel threshold-based channel switching system, called balanced channel control system. In our design, the threshold could be dynamically deduced according to the real-time traffic and corresponding throughput of each node. Our threshold control model is based on a fuzzy logic control. We also designed a novel dynamic channel assignment scheme, which is used for the selection of new channel. To perform channel switch between a pair of neighboring nodes, we designed a channel switch scheduler. The channel switch scheduler is used to perform channel-switch processing for sender and receiver over enhanced routing protocols. We evaluated this system in NS2, and the simulation results showed that our BCS improves the throughput by 12.3%-72.8% and reduces the latency by 23.2%-52.3% over existing solutions. Our work is not confined with channel switch over a single link, it can be extended to ensure stability over a local region,
and hence, the network as a whole. It is well known that interference in reality is very complicated. We plan to conduct real system experiments with our balanced control system. Detailed study of this extension will be our future work.

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Xiaoguang Li received Master Degree in Computer Science from Athlone Institute of Technology in 2009; and Bachelor of Science in Information and Computational Science from Nanjing University of Science and Technology in 2004. She currently works toward the Ph.D. degree in the Department of Computer and Information Sciences at Temple University, under the supervision of Dr. Jie Wu. Her current research interests mainly focus on wireless sensor networks, delay tolerant networks and wireless mesh networks.

Jie Wu is the chair and a professor in the Department of Computer and Information Sciences, Temple University. Prior to joining Temple University, he was a program director at National Science Foundation. His research interests include wireless networks and mobile computing, routing protocols, fault-tolerant computing, and interconnection networks. He has published more than 550 papers in various journals and conference proceedings. He serves in the editorial board of the IEEE Transactions on Computers, IEEE Transactions on Mobile Computing, and Journal of Parallel and Distributed Computing. Dr. Wu is program co-chair for IEEE INFOCOM 2011. He was also general co-chair for IEEE MASS 2006, IEEE IPDPS 2008, ACM WiMD 2009, and IEEE/ACM DCOSS 2009. He also served as panel chair for ACM MobiCom 2009. He has served as an IEEE computer society distinguished visitor. Currently, he is the chair of the IEEE Technical Committee on Distributed Processing (TCDP) and ACM distinguished speaker. Dr. Wu is a Fellow of the IEEE.

Shan Lin received his B. E. degree in computer science and engineering from Shanghai Jiao Tong University in 2004, and his PhD degree in Computer Science from the University of Virginia in 2010. He
joined Temple University as a Tenure Track Assistant Professor in 2010. He was awarded as the SAIC scholar in 2009. Lin's primary research interests are in the areas of cyber physical systems, networked embedded systems, and wireless sensor networks. He has been investigating feedback control based approaches to networked system designs, including wireless networking and system composition. He is also involved in building wireless sensing systems for pervasive medical care and fire fighting. His major papers have been published in ACM MobiSys, ACM SenSys, IEEE RTSS, IEEE INFOCOM, ACM TECS, and ACM IPSN. He is a member of IEEE and ACM.

Xiaojiang (James) Du is currently an associate professor in the Department of Computer and Information Sciences at Temple University. Dr. Du received the B.S. degree in electrical engineering from Tsinghua University, Beijing, China in 1996, and the MS and Ph.D. degree in electrical engineering from the University of Maryland College Park in 2002 and 2003, respectively. Dr. Du was an Assistant Professor in the Department of Computer Science at North Dakota State University between August 2004 and July 2009, where he received the Excellence in Research Award in May 2009. His research interests are security, wireless networks, computer networks and systems. He has published over 100 journal and conference papers in these areas, and has been awarded more than $2M research grants from the US National Science Foundation (NSF) and Army Research Office. He serves on the editorial boards of four international journals. Dr. Du is the Chair of the Computer and Network Security Symposium of the IEEE/ACM International Wireless Communication and Mobile Computing conference 2006 - 2010. He is a Technical Program Committee (TPC) member of several premier ACM/IEEE conferences such as INFOCOM (2007 - 2012), IM, NOMS, ICC, GLOBECOM, WCNC, BroadNet, and IPCC. Dr. Du is a Senior Member of IEEE and a Life Member of ACM.