國立交通大學

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碩士論文

一個無線區域網路 EDCA 上支援媒體串流 之無接縫交遞機制 A Seamless Handoff Scheme Supporting Media

Streaming on EDCA Over Wireless LAN

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中華民國九十五年八月

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摘要

支援無線區域網路行動性的無接縫及快速交遞機制的部份研究中,均以交遞 發生前先行偵測交遞對象為基礎,並已發展完整。「鄰近圖」(neighbor graph)是 偵測交遞對象的方法中最常見的一種。但「鄰近圖」因為先天特性的一些缺點, 至今都還沒被廣泛的應用。在這篇論文中,我們提出了所謂「不連續掃描」的機 制來承襲「鄰近圖」的功能且排除其不實用的缺點。在「不連續掃描」的基礎上, 我們更進一步提出了一些機制來幫助行動工作站在交遞前選擇合適的基地臺作 為交遞對象,並用模型分析及模擬的方法來驗證這些機制的效能。「不連續掃描」 最重要之疑慮可能在於對行動工作站的服務品質產生衝擊。然而,模擬結果顯 示,對一個在無線區域網路 EDCA 上運作的媒體串流而言,「不連續掃描」造 成的中斷將可以被控制在可接受的程度以內。

A Seamless Handoff Scheme Supporting Media Streaming on EDCA Over Wireless LAN

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Abstract

Seamless or fast handoff schemes to support mobility of IEEE 802.11 Wireless LANs have been well developed based on the assumption that a set of next-AP/AR candidates are available prior to a handoff. A neighbor graph is one of the major strategies used to provide such information among the related studies. However, neighbor graphs have not been widely applied yet due to its inherent drawbacks. In this thesis, a so-called "discrete scan" scheme is proposed to substitute the function of a neighbor graph without drawbacks. Moreover, several mechanisms based on discrete scan are presented to help a mobile node select a desired next AP to handoff to. The analytical model and simulation are elaborated to show performance of the approaches. Discrete scan may bring concerns on its impact to received QoS of the mobile node. However, the simulation results show that disruptions caused by discrete scan can be controlled within an acceptable value for a media streaming device working under EDCA over wireless LAN.

Keywords: Neighbor graph, handoff, discrete scan, media streaming, EDCA

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摘要		i
Abstract	t	ii
Acknowl	vledgement	iii
List of Fi	Figures	vi
List of Ta	۲ables	vi
Chapter	r 1 Introduction	1
Chapter	r 2 Related works	7
2.1	Review of Layer-2 And Layer-3 Handoffs	8
2.1.1	IEEE 802.11 Handoffs	8
2.1.2	Mobile IP Handoffs	9
2.2	Fast Handoff Schemes Based on Next AP/FA Prediction	12
2.2.1	AP Probe Latency	
2.2.2	Association Latency	12
2.2.3	802.1x Authentication Latency	13
2.2.4	Layer-3 Handoff Latency	13
2.2.5	Cross-Layer Design	13
2.3	Summary of Related Works	14
Chapter	r 3 Proposed Approaches	15
3.1	Discrete Scan Scheme	15
3.2	Collection of Next-AP candidates	18
3.3	Mechanisms to Select a Next AP	21
3.3.1	Mechanisms to Select the Nearest AP	21
3.3.2	Setting of a Sniffing Period	25
3.3.3	Performance of Station Count Mechanism	30

Table of Contents

3.3.4	Select Best Next AP Instead of the Nearest AP	35
3.4	Impact on Service QoS	36
Chapter	4 Performance Evaluation	42
4.1	Simulation Environment	42
4.2	Station Count Mechanism in EDCA Environment	44
4.2.1	Setting of a Sniffing Period in EDCA Environment	44
4.2.2	Performance of Station Count Mechanism in EDCA Environment	47
4.3	Impact on Service QoS in EDCA Environment	51
Chapter	5 Conclusion and Future Works	53
Referenc	es	56



List of Figures

Figure 1.1 Set of neighboring APs and set of next-AP candidates by discrete scan	6
Figure 2.1 Overall latency in a layer-2 and layer-3 handoff	11
Figure 3.1 Scheme for wireless NIC in pre-handoff periods	17
Figure 3.2 Three kinds of evidences to discover an AP	18
Figure 3.3 Example sniffing scheme in the environment of 3 channels	20
Figure 3.4 the higher station distribution ratio indicates the nearer AP	22
Figure 3.5 The station counts in two BSSs	24
Figure 3.6 Definition of T_s and T_c , slot time of transmission with success and collision	27
Figure 3.7 Transient number of stations (r) v.s. number of stations (n) for different T_{sniff} values	29
Figure 3.8 Scenario for performance estimation	32
Figure 3.9 Hit ratios v.s. distances in balanced number of stations	33
Figure 3.10 Variation of hit ratios with imbalance of number of stations	34
Figure 3.11 Total disruptions induced with a discrete scan	38
Figure 3.12 Probability to transmit a frame in 10 ms v.s. number of stations in a BSS	41
Figure 4.1 Simulation configuration	43
Figure 4.2 Number of transient stations v.s. number of stations in EDCA of 70% TCP and 30% Vo	эIР
nodes	45
Figure 4.3 Number of transient stations v.s. number of stations in EDCA with 100% TCP nodes	46
Figure 4.4 Hit ratios v.s. distances in balanced number of stations in EDCA WLAN of 30% Vo	эIР
nodes and 70% TCP nodes.	47
Figure 4.5 Hit ratios v.s. distances in balanced number of stationsin EDCA WLAN of 100% Te	СР
traffic	48
Figure 4.6 Hit ratios v.s. imbalanced number of stations in EDCA of 30% VoIP nodes and 70% To	СР
nodes	49
Figure 4.7 Hit ratios v.s. imbalance of number of stations in EDCA of 100% TCP traffic	50
Figure 4.8 Probability to send a frame in 10 ms v.s. number of stations in 11 Mbps EDCA wirele	ess
LAN	52

List of Tables

Table 3.1 Parameters for IEEE 802.11b wireless LAN	
Table 4.1 Default values of parameters for original EDCA.	44

Chapter 1 Introduction

In recent years, wireless LANs have been widely deployed for public mobile Internet services. Public wireless LANs can provide high-speed Internet connectivity using portable devices such as laptop computers and personal digital assistants (PDA). An emerging application to wireless LAN system could be Voice over IP (VoIP), which requires networks to support real-time transmission such as bounded latency and jitter. One of the key issues for VoIP application to support real time servise is providing the Quality of Service (QoS) that meets the user expectations. This is recognized as the single biggest challenge in providing high-quality voice on wired IP-based networks. In wireless networks, one of the major issues affecting QoS is to minimize the service disruption during handoffs of the mobile nodes.

It is recognized that an intra-subnet handoff across access points (APs) in an administrative domain, so-called layer-2 handoff, normally brings a service disruption of several hundreds of mini-seconds (ms), and an inter-subnet handoff across different domains, so-called layer-3 handoff, would be disrupted in periods of several seconds. However, to meet the requirement of real-time services, the maximum latency would be limited to less than 100 ms. Therefore, lots of investigation devoted to minimizing the handoff latency, which is defined as the period while a mobile node is unable to receive application traffic due to the handoff.

Among the researches focusing on minimization handoff latency, schemes based on prediction of next AP for a layer-2 handoff, or next access router (AR) for a layer-3 handoff are proposed in lots of literatures. Neighbor graphs, which are defined as a specific data structure that stores information of set of neighboring APs or/and ARs with respect to current AP or/and AR, are one of major approaches to predict next APs or/and ARs to handoff to. With aids of neighbor graphs, mobile nodes in a layer-2 handoff will probe the channels in which at the least one of the neighboring APs exists. Moreover, mobile nodes can terminate probing the channel immediately after the last neighboring AP has responded rather than wait until maximum channel time. Therefore, the probe phase in layer-2 handoffs can be significantly shortened with the help of neighbor graphs [1].

The reassociation latency in layer-2 handoffs can be reduced an order by the scheme that the current AP caches secure information of mobile nodes to all of the APs in neighbor graphs prior to occurrence of layer-2 handoffs [2]. The 802.1x reauthentication to new AP would also bring significant latency to layer-2 handoffs. By means of proactive key distribution to all of the APs in neighbor graphs are reported to effectively eliminate the 802.1x reauthentication latency from 800 ms up to 10 ms [3].

The layer-3 latency can be effectively reduced to a level of layer-2 handoffs with aids of neighbor graphs. The neighbor-casting scheme sets up temporary links from current AR to all ARs in neighbor graphs in order to maintain the connection to the mobile node via the relay of current AP to next AP in a layer-3 handoff. Therefore, the latency induced by a layer-3 handoff is eventually reduced to the same level as disruptions caused by the layer-2 handoff [4].

The neighbor graphs include information of all possible APs/ARs that a mobile may handoff to from the current AP/AR. The size of a neighbor graph (i.e., number of APs/ARs in a neighbor graph) may be large enough to cause severe overhead to wired networks. Size of a neighbor graph reflects the number of potential APs to be probed for a mobile node to complete the probe phase in a layer-2 fast handoff. In both context caching and neighbor-casting schemes, the context of the mobile node and the arriving packets for the mobile node during a layer-3 handoff have to be duplicated as

many copies as the number of APs/ARs in neighbor graphs. Therefore, the size of a neighbor graph stands for the overhead to wired bandwidth in a fast handoff scheme. In proactive key distribution scheme, the pairwise master keys (PMK) are generated and distributed to all APs in neighbor graphs. The size of a neighbor graph thus responds to the overhead and the security risk caused by abuse of key distribution.

To reduce the size of a neighbor graph, some schemes such as selective neighbor caching [5] and frequent handoff region (FHR) [6] are proposed. Moreover, global positioning system (GPS) is proposed to update the topology information of a mobile node dynamically for the sake of predicting the next AP among APs, in a neighbor graph before a handoff is initiated. Based on the assumption that next AP and next AR are known prior to the occurrence of a handoff with aids of a neighbor graph and real time topology information, a cross-layer fast handoff scheme is proposed to minimize overall latency for a layer-2 and layer-3 handoff to less than 50 ms in the experiment results [7].

It is generally recommended to configure a neighbor graph with a distributed manner such that each AP/AR stores and maintains its own neighbor graph. The neighbor graph can be automatically learned by the individual AP/AR with the reassociation frames and registration packets from various mobile nodes. Distributed neighbor graphs require the supports from neighboring APs/ARs as well as the mobile nodes roaming through the coverage of APs/ARs.

In fact, not all of the mobile nodes have to invoke for seamless handoffs. For example, a laptop with a wireless network interface card (NIC) may be considered as a mobile node since it may be carried from the coverage area of an AP/AR to another. With its volume and weight, the user may seldom play real time application (such as VoIP) while the laptop is being carried across different wireless LAN. Therefore, seamless handoffs may be not a mandatory function for a laptop with wireless NIC unless the notebook is especially designed for used in a vehicle. A hand carried device such as personal digital assistants (PDA) may run only in non-real-time applications, such as web browser or specific remote database queries. A hand held device for non-real-time applications may neither need feature of seamless handoffs since disruptions of few hundreds mini-seconds is generally tolerable to users. A typical example that really invokes for seamless handoffs is recognized as media streaming, such as a WiFi VoIP, with both attributes of hand held devices and real-time application. Since only particular clients request for seamless handoffs, neighbor graph scheme may induce unnecessary overhead by non-seamless-handoff nodes because neighbor graph scheme is implemented in AP that provides universal service to all of its clients. For example, a stationary station may work as a client of an AR **WWW** with neighbor graph function. By ignoring the non-seamless-handoff attribute, an AR may buffer and forward the packets to all its neighboring ARs in case a sudden termination from the client is misinterpreted as an event of handoff, so that misinterpretation as well as bandwidth waste may occur.

Millions of IEEE 802.11 APs/ARs as well as mobile nodes have been deployed without supports to neighbor graphs. One of the reasons for the popularity of IEEE 802.11 wireless LAN is due to the features of simple configuration and low cost. To support neighbor graph mechanism, unfortunately, increases complexity and cost to re-configure an IEEE 802.11 wireless LAN. Moreover, a new protocol will be required to enable cooperation among APs/ARs and mobile nodes though some of mobile nodes do not require seamless handoffs from wireless circumstance.

The factors described above somehow explain the difficulty for neighbor graph mechanism to be implemented in existed wireless networks, though researches have proven with experiments that seamless inter-subnet handoffs can be eventually achieved with integral fast handoff schemes benefit by neighbor graph mechanism. In this thesis, the so-called "discrete scan", a simple and effective scheme to collect the set of potential next APs in a layer-2 handoff is proposed. A mobile node with discrete scan scheme executes the process of passive scan a certain time ahead of handoff initiation. To avoid disruptions lasting longer than 50 ms, process of passive scan is divided into smaller pieces of sniffing periods; therefore, the probe delay in layer-2 handoff is decomposed into pieces of disruption that users are unable to sense. Discrete scan is designed to implement in a media streaming application such as WiFi VoIP that typically desires function of seamless handoffs. Benefit by features of dedicated hardware and relative low working bit rate (usually less than 100 Kbps bidirectional) compared with its network interface card (NIC) (11Mbps), discrete scan scheme can take place of neighbor graphs without collaboration from APs as well as other mobile nodes.

The characteristic that a mobile node spends its own effort to predict the next AP to fasten its handoff may allow the seamless handoff schemes to utilize to discrete scan. Without any support from AP, the mobile nodes using discrete scan can perform a fast handoff by elimination of probe latency from a layer-2 handoff. The confusion for an AP to provide a seamless handoff service to a node without such demand can be avoided because discrete scan will be only implemented in the nodes that may invoke seamless handoffs.

Compared with neighbor graphs, the set size of candidate next-APs collected by discrete scan scheme is much smaller than that by neighbor graphs. As shown in Figure 1.1, discrete scan scheme discovers only a subset of neighboring APs of which signals can reach to the mobile node. Taking advantage of examined information from frames received in discrete scan, such as received signal strength (RSS), MAC address and network allocation vector (NAV), several mechanisms to assist the mobile node to select the nearest or/and the best AP among the candidates is introduced in this thesis.

With discrete scan as well as its auluxiary mechansims, the mobile node is able to select the most appropriate AP to handoff to before handoff is initiated. Thus, latency induced by probe phase in a layer-2 handoff is significantly reduced to a ProbeRequest from the mobile node and a ProbeResponse from the selected AP, without modification to the existed infrastructure of IEEE 802.11 wireless LANs. Since the latency spent in probe phase is recognized as more than 90% of overall layer-2 handoff latency, discrete scan with its auluxiary mechansims expect to reduce 90% of latency in a layer-2 handoff. With such significant improvement but without modification required to the current system, the realization of discrete scan scheme for a mobile node is reasonably much easier than that of neighbor graph scheme.



Figure 1.1 Set of neighboring APs and set of next-AP candidates in discrete scan.

Discrete scan is supposed to support seamless layer-3 handoffs with the fast handoff schemes proposed for neighbor graphs. Substituting for the functions of a neighbor graph and its extensions, discrete scan with auxiliary mechanisms proposed in this thesis helps a mobile node determine the next AP in advance of the trigger of a handoff. The only insufficiency for discrete scan scheme to take place of a neighbor graph lays on the lack of the ability to discover next AR because discrete scan is absolutely designed for layer-2 operation. However, a lookup table built-in the current AP, that outputs next AR with input of the next AP, will simply address the problem.

One of major concerns on discrete scan scheme is its impact on the service QoS received in the periods of discrete scan, because discrete scan decomposes the latency of probe phase in a layer-2 handoff into pieces. Benefit by the high transmission priority of VoIP traffic in wireless EDCA environment, the simulation results in this thesis show that QoS degradation due to discrete scan is controlled within a tolerable value for a real-time application.

The rest of this thesis is organized as follows. Chapter 2 introduces the background of latency in layer-2 and layer-3 handoffs. The relative works involved in neighbor graphs are introduced briefly in this chapter as well. In Chapter 3, we elaborate discrete scan scheme and the mechanisms to select next AP. Besides, in Chapter 3, the performance and impact on service QoS are evaluated by numerical results based on analytical model of DCF wireless LAN environment, and those in EDCA environment are demonstrated with simulation results in Chapter 4. Finally, the conclusion and future works are presented in Chapter 5.

Chapter 2 Related works

In this chapter, we first briefly review the processes of a layer-2 handoff and a layer-3 handoff as those extracted from [7]. IEEE 802.11 wireless LAN and mobile IP (MIP) standard are interpreted as typical samples for a layer-2 handoff as well as a layer-3 handoff.

Next to review on handoff process, the breakdown latency for a layer-2 and a layer-3 handoff is presented as those proposed in [7]. References in this section show that each division for handoff latency can be significantly reduced based on the aids of neighbor graphs that provide current AP/AR and the mobile node a set of neighboring APs/ARs that contains the next AP/AR in a handoff.

Finally we conclude that the capability to predict next APs/ARs in advance of a handoff is the key point to fulfill a fast handoff. A new scheme for a mobile node to predict or select the next AP/FA in a handoff is proposed in this thesis to replace the function of a neighbor graph. The proposed scheme in this thesis will be able to work with the strategies designed for working with a neighbor graph. A series of fast handoff studies based on the existence of a neighbor graph are addressed in section 2.2.

2.1 Review of Layer-2 And Layer-3 Handoffs

2.1.1 IEEE 802.11 Handoffs

A layer-2 handoff consists of three phases: probe, authentication, and reassociation. In the probe phase, a mobile node discovers next-AP candidates

through either an active or a passive scan. In an active scan, a mobile node broadcasts through a selected channel a ProbeRequest message with a particular Service Set Identifier (SSID). If the SSID matches an AP's configuration, then the AP responds with a ProbeResponse to the mobile node, and the mobile node can therefore be made aware of the presence of the AP. If the mobile node instead uses a passive scan, then it does not issue any message but listens to Beacon messages broadcast periodically by APs on channels of interest.

With AP information obtained from the ProbeResponse or Beacon message, the mobile node selects a new AP to camp on based on the measure of received signal strengths. Following the probe phase, the mobile node performs 802.11 authentication (open system or WEP), and then reassociation phases with the newly selected AP. In the authentication phase, the mobile node exchanges 802.11 authentication messages with the AP. In the reassociation phase, the mobile node sends a ReassociationRequest to the AP and receives a ReassociationResponse replied by the AP. The receipt of the last message terminates the 802.11 handoff process.

As a port-based network access protocol, IEEE 802.1x provides authentication and key management under various 802 LAN infrastructures, and is now extensively adopted in 802.11 WLANs to resolve the limitations of WEP. An 802.1x-enabled AP acts as an authenticator controlling the mobile node's access to the Internet. The authenticator communicates with an authentication server that makes authorization decision on the access requests sent by a mobile node (called a supplicant in 802.1x terms). Either the mobile node or the authenticator may initiate an 802.1x authentication immediately after the reassociation phase is completed. If the authentication is successful, then the authentication server sends a pair-wise master key (PMK) to the authenticator, which then initiates an 802.11i four-way handshake procedure to synchronize the PMK with the mobile node and to generate pair-wise temporal keys (PTKs). The 802.1x control port of the authenticator is then unblocked for the mobile node, and the mobile node can then send and receive messages protected by the PTKs.

2.1.2 Mobile IP Handoffs

Mobile IP (MIP) is an Internet standards track protocol that enhances the existing IP protocol to accommodate host mobility. In MIP, a special host called a mobility agent (MA) maintains registration information for mobile nodes. When a mobile node moves away from its home network, the MA located in the mobile node's home network, called the mobile node's home agent (HA), tunnels packets for the mobile node. A mobile node away from its home network can retain its connection to the Internet aided by the HA and the FA.

MIP facilitates care-of address (CoA) to identify a mobile node in the visited network. As a mobile node registers a CoA with the mobile node's HA, then the FA intercepts all tunneled packets destined for the mobile node, and delivers then to the mobile node. When a mobile node detects that the current FA (cFA) is no longer accessible, it initiates a layer-3 handoff from the cFA to the next FA (nFA). A layer-3 handoff consists of two phases: the mobile node must first discover the nFA and then register with the mobile node's HA through the nFA, as described below:

a. Agent Discovery: This concerns how a mobile node becomes aware of the presence of an nFA. Every MA can be uniquely identified by its AgentAdvertisement message. A mobile node may either passively listen to AgentAdvertisement messages broadcasted periodically by the nFA, or actively issue an AgentSolicitation message to request an advertisement. b. Registration: This informs the HA of a mobile node's CoA. The mobile node issues a RegistrationRequest message to the nFA, from which the message is then forwarded to the HA. The HA sends a RegistrationReply to the mobile node to confirm the registration with relay of the nFA.

The process through which a mobile node detects that a cFA is no longer accessible is called move detection. MIP specifies two move-detection principles: the advertisement expiration and the network prefix change. Each AgentAdvertisement in MIP carries an advertisement lifetime. If the lifetime of the most recently received advertisement expires, then the mobile node may assume that the cFA is unreachable, which generally leads to long move detection delays, as MIP suggests that the advertisement lifetime should be long enough to tolerate three consecutive losses of advertisements. Alternatively, if the mobile node receives an AgentAdvertisement with a network prefix different from that of the mobile node's current CoA, then the mobile node may deduce that cFA is unreachable, leading to a long move-detection delay as the mobile node can receive nFA's advertisement only after a layer-2 handoff.

2.2 Fast Handoff Schemes Based on Next AP/FA Prediction





Figure 2.1 illustrates total latency incurred in layer-2 and layer-3 handoffs. Many studies have attempted to reduce the delay in different activity sections to shorten handoff latency, as described in following subsections.

2.2.1 AP Probe Latency

In [1], Mishra et al. have noted with experiments that AP probe phase latency significantly contributes to the layer-2 handoff latency, and recommended using neighbor graphs to eliminate the delay in AP probe. The mobile node only needs to probe the neighboring APs with respect to the current AP. An AP is a neighbor of another AP only if a handoff from the latter to the former has occurred recently. Neighbor graphs thus dynamically capture temporal handoff-to relationships.



2.2.2 Association Latency

Mishra et al. [2] have also facilitated neighbor graphs to lower the reassociation latency by proactively caching security information to neighboring APs, where security information is needed to establish secure communication channels between APs. The experiment results show reassociation latency could be reduced from 15.37 ms to 1.69 ms if the mobile node's context has been cached to the next AP prior to a handoff. To reduce the costly signaling overhead brought by proactive neighbor casting, Sangheon Pack [5] recommended only the neighboring APs with relative high probabilities that a mobile node may handoff to be selected as next-AP candidates.

2.2.3 802.1x Authentication Latency

Exploiting a neighbor graph as well as proactive key distribution to candidate set of next APs with which a mobile node may reassociate, the 802.1x reauthentication latency incurring in layer-2 handoffs is reduced down to 50 ms with comparison to 800 ms for a full 802.1x authentication [3].

2.2.4 Layer-3 Handoff Latency

Several schemes such as neighbor casting [4], centralized decision engines [9], and Frequent Handoff Region [6] have been proposed to collect sets of next-FA candidates. In the neighbor-casting scheme, the current FA sets tunnels to all of the next-FA candidates for the purpose to relay packets for a mobile node after it losses its connection to the mobile node caused by a layer-3 handoff. With such a mechanism that the cFA buffers-and-forwards packets to nFA for the mobile node, the latency brought with a layer-3 handoff is eliminated to the level as a layer-2 handoff [8].

2.2.5 Cross-Layer Design

It is proposed to take advantage of neighbor graphs and topology information offered by auxiliary equipment such as Global Position System (GPS) to derive an accurate prediction on next FA in a layer-3 handoff. Move detection and agent discovery delay in a layer-3 handoff are minimized by post handoff trigger scheme because next FA has been known before the handoff. Further, registration for the predicted next FA to HA could be done before layer-3 handoff occurs, called pre-handoff trigger scheme. Therefore, the total latency for layer-3 handoffs is eliminated to its minimum with a cross-layer design fast handoff scheme in [7]. Experiment results in [7] show that cross-layer design with the supports of accurate prediction of the next AP reduces overall end-to-end layer-3 handoff latency up to 50 ms so that a seamless layer-3 handoff can be achieved.

2.3 Summary of Related Works

A series of fast handoff studies have been developed based on the hypothesis that the information of the next APs/FAs is collected or predicted to somewhat degree by a particular mechanism. The neighbor graph [1] [2] [3] [4] [7] is one of the major mechanisms utilized to provide information of a set of the next-AP/FA candidates. To further improve accuracy of prediction or reduce the number of next-AP candidates, some enhancement such as design engine [5] and frequent handoff region [6] are presented. The ultimate prediction mechanism that aims on the ability to detect exactly the next AP or/and AR prior to handoffs initiated was proposed to take advantage of a neighbor graph with aids of topology information updated by a GPS in the mobile node [7]. It concludes that the support to know the next AP/AR in an impending handoff is the key to minimize an overall latency, because a series of fast handoff schemes based on known next AP/AR have been presented with experiment results.

Chapter 3 Proposed Approaches

In this chapter, we propose a so-called "discrete scan" scheme and relative auxiliary mechanisms that collect a set of candidates for next-APs and eventually select the next AP for a mobile node before a forthcoming handoff. Moreover, various existed mechanisms may select next AP with different properties of the next-AP candidates, such as the nearest AP, the approached AP, the AP of highest available bandwidth, or their combination.

Compared with neighbor graphs, discrete scan scheme provides a set of next-AP candidates as a subset of neighboring APs. With the help of proposed auxiliary mechanisms, discrete scan selects an appropriate next AP for a mobile node to handoff to, taking the place of the function of a neighbor graphs with aids of topological information by GPS system proposed in [7].

As a cost, discrete scan scheme contributes an impact to service QoS received by the mobile node in a "pre-handoff" period that is defined as the duration from initiating the discrete scan scheme to the moment when a handoff is initiated. However, taking the advantages of high transmission priority and relative low bit rate for a VoIP connection in EDCA wireless LANs, the degradation may be limited to a degree that the users may tolerate or even may ignore. Estimation on the QoS degradation with analytical approach and simulation results presented in this and next chapter shows that the disruptions caused by discrete scan are bounded to 50 ms based on certain practical assumption.

3.1 Discrete Scan Scheme

The main idea of discrete scan is to allow a mobile node utilizing idle time of its wireless NIC to survey its wireless environment in the periods about to handoff. As described in Chapter 1, only hand carried sets in real-time application, typically as WiFi phones, in fact require seamless handoffs in IEEE 802.11 wireless environment. A hand held device normally implies a dedicated hardware that allows modification of firmware and even its hardware to meet particular requirements. Real-time applications usually work in relative low bit rates with respect to the available bandwidth of its wireless NIC. Taking a WiFi VoIP service as a example: the WiFi phones operate with bidirectional traffic both with 64Kbps constant bit rate (CBR) in an IEEE 802.11b wireless LAN which supports at the least 2 Mbps half duplex bandwidth. From the viewpoint of the wireless NIC of a WiFi phone, traffic transmission and receiving time takes about 64K*2 / 2M = 6.4% of total operation time. More time may be taken by transmission overhead such as physical layer overhead, MAC header, MAC layer control frames, waiting time for media contention, retransmission due to collisions, etc, however, NIC keeps in idle status for a significant portion of time.

In discrete scan, a mobile node, such as a WiFi phone set, will make use of the idle time in pre-handoff duration to sniff its wireless environment to collect a set of next-AP candidates for a forthcoming handoff. A pre-handoff period is initiated with an event that received signal strength (RSS) from current AP is detected to be lower than a preset threshold δ_2 for a preset period. Obviously, threshold δ_2 shall be set a little bit larger than the threshold δ_1 , the threshold RSS to initiate a layer-2 handoff in IEEE 802.11.

In the pre-handoff periods, the wireless NIC of a mobile node operates in a mode that switches channels with specific scheme. Figure 3.1 below illustrates the typical scheme for the wireless NIC in the pre-handoff periods. While working in pre-handoff mode, the wireless NIC has to return to working channel to maintain VoIP connection alive with a certain level of QoS. The time between two working periods, that is, the time of sniffing period plus two switching periods, should not be longer than β ms so that the maximum latency allowed by a real time application, α ms, is maintained if the node will get at least two times of transmission within (α – β) ms, as shown in Figure 3.1. During working periods, bidirectional traffic transmits between current AP and the mobile node at least once to deliver the traffics generated in previous absence from working channel. By the end of working periods, the mobile node should issue an extra control frame of power save mode (PSM) to notice the current AP to suspend the packets for the mobile node.





Sniffing periods: The periods for sniffing channels (channel 1~10)

Figure 3.1 Scheme for wireless NIC in pre-handoff periods

In principle, discrete scan amortizes the latency for probe phase in a layer-2 handoff with pieces of sniffing periods in a pre-handoff duration. Each sniffing period should be short enough in order to maintain the induced disruption within tolerable limits for a real time application. The setting of the length of sniffing periods will sometimes become a trade-off between efficiency of collecting next-AP candidates and degradation of service QoS in a pre-handoff period.

3.2 Collection of Next-AP Candidates

To substitute for the function of neighbor graphs, discrete scan must be able to collect a set of next-AP candidates. One of the natures particular to a seamless handoff is that a mobile node must have entered the coverage area of its next AP before a layer-2 handoff is triggered. Therefore, a mobile node can discover the entire set of next-AP candidates by means of sniffing frames transmitted from APs in various channels. Note that the number of next-AP candidates discovered by discrete scan scheme is much less than the number of neighboring APs because only those neighboring APs located at the area that the mobile node can listen to can be found (refer to Figure 1.1).

In Figure 3.2, there are three kinds of evidences to identify the discovery of a next-AP candidate, as described in the following:



Figure 3.2 Three kinds of evidences to discover an AP

- a. A complete transmission with both of data frame and corresponding ACK frame is sniffed. In this case, it is inferred that a next-AP candidate and one of its clients locates within the area the mobile node can listen to. In case none of active stations locate in the coverage of the mobile node, the AP will not be found under this condition.
- b. Two frames sent from a node (of the same MAC address) to different destinations are sniffed. By the nature of infrastructure wireless LAN, only the AP of a basic service set (BSS) can send frames to different destinations, so that the sender of two frames to distinct receivers will be identified as an AP. Since the frames from AP have been sniffed, the mobile node must be locating within the coverage of the AP, therefore, the AP is recognized as one of candidate next AP. If a BSS has less than two active stations in sniffing periods, the next-AP candidate will not be found under this condition.
- c. A beacon is sniffed, same as a traditional scheme to identify an AP. Because a sniffing period is shorter than a beacon interval, there is no guarantee that all of next-AP candidates will be found in a sniffing period. However, utilizing the convention that all the beacon intervals are typically set as 100 ms, each beacon interval may be considered as multiple time slots with each slot equal to a sniffing period. A mobile node may discover all of its next-AP candidates in different channels through their beacons, in case it separately sniffs all the time slots for each channel.

An example design for the sniffing scheme is illustrated in Figure 3.3. A sniffing period is designated by 20 ms for each interval of 60 ms. Assume that only three channels are deployed in the concerned area. A beacon interval is consisted of 5 slots, from slot 0 to slot 4, with each slot time 20 ms, and two channels that are different

from the working channel will be listened in a sequence of sniffed periods. The sniffed channels should be interleaved so that each channel is probed evenly.



The maximal time required for a mobile node to discover all APs through beacons should be taken into account to set the value of δ_2 , the threshold RSS to start a pre-handoff period. As an example, 20 ms slot of each 60 ms interval is used for sniffing, that is, 1/3 of total run time is shared to sniff for the next-AP candidates in all channels except the working channel. Each scanned channel takes 5 sniffing periods, equal to 100 ms in total, to ensure to discover of all of the next-AP candidates active in the channel. Therefore, it takes (number of channels–1) * 3 * 100 ms to complete the discrete scan scheme on all of channels. With at the most 11 channels in IEEE 802.11b wireless LAN environment, thus, it takes (11-1) * 3 * 100 ms, equal to 3 sec, to complete a full passive scan. Therefore, a pre-handoff period is suggested to be 3 seconds ahead of a layer-2 handoff.

3.3 Mechanisms to Select a Next AP

3.3.1 Mechanisms to Select the Nearest AP

Discrete scan surpasses the neighbor graph in the ability to select a much suitable next AP for a mobile node before the handoff starts. Taking the advantage of the information extracted from MAC frame headers that a mobile node listens through its discrete scan scheme; a desired next AP can be determined by the mobile node along with extra supports from existed wireless LAN system. The feature to determine the next AP may take over the function of a neighbor graph with aids of GPS as proposed in [7].

The nearest AP is normally chosen as the next AP in a handoff because of property of the best signal strength. In a conventional handoff schemes, a mobile node determines the nearest AP by the RSS received from beacons or frames sent by APs during the probe phase of a layer-2 handoff. However, it is recognized that the strongest RSS may be not a prefect indicator of the nearest AP because of the effects by multi-paths interference of microwave communications. Nevertheless, the principle of choosing the next AP by indication of RSS is widely implemented in wireless NIC, since no better indicator is available so far.

Discrete scan, discovering next-AP candidates through frames or beacons sent by APs, may restrict the capability to select next AP by their RSS. Furthermore, several mechanisms of a new idea with aids of MAC layer information are proposed in this section to compensate the insufficiency of RSS scheme currently supported by physical layer.

A mobile node with discrete scan may derive information to indicate the next AP of the handoff by the MAC headers in the frames listened in its sniffing periods. By

reading MAC layer headers, a mechanism is designed to record the number of stations that can be listened by the mobile node in each channel. A mobile node learns a station with its MAC address in the frame header transmitted from the station.

The first proposed mechanism for a mobile node to estimate its distances from candidate APs is named "station distribution ratio" that is defined as the ratio of the numbers of stations of in the coverage of the mobile node to all of the stations in a BSS. As illustrated in Figure 3.4(a) (b), the higher station distribution ratio is a BSS, the nearer is the AP to the mobile node.



Figure 3.4 the higher station distribution ratio indicates the nearer AP.

The key mechanism of station distribution ratio lies in how a mobile node distinguishes a station within its area from those out of its area in a BSS. By reviewing the Figure 3.2(a) once more, a mobile node can identify a station in its coverage area if the mobile node can receive both a data frame and the corresponding ACK frame in its discrete scan scheme. Similarly, if a mobile node receives only one of a data frame or the corresponding ACK frame, as illustrated in Figure 3.2 (b), the mobile node infers that the station in the BSS locates out of its coverage.

Another good feature that discrete scan surpasses neighbor graphs is its slight

penalty if an incorrect AP is selected by the auxiliary mechanisms. A mobile node always locates at the overlapped coverage of all of the next-AP candidates discovered in the discrete scan scheme, therefore, an improper selection among the next-AP candidates will not lead to a failure of a handoff. On the contrary, a mobile node will lose connection if the GPS in a mobile node predicts a neighboring AP that can not reach to the mobile node as the next AP.

As illustrated in Figure 3.4(c), the cluster of stations in a BSS may lead the station distribution ratio mechanism to an improper selection of the next AP. However, the penalty for a mobile node to handoff to a farer AP may just cause another handoff to occur earlier.

There are some shortcomings of the station distribution ratio mechanism. The station distribution ratio mechanism will be active only after an AP is discovered because a mobile node infers a station out of its coverage by frames from AP instead of the station. Consequently, the station distribution ratio mechanism requires a certain amount of time to evaluate an AP after it is discovered. Besides, a mobile node needs more extra cache to memorize the MAC addresses of discovered stations to avoid counting a station twice.

To overcome the drawbacks described above, the station distribution ratio mechanism is simplified and submitted as the second mechanism for a mobile node to select the nearest AP among candidate APs. The mechanism is named as "station count" because it suggests a mobile node to select an AP with the largest number of stations counted by a mobile node in the last sniffing period prior to a handoff. A mobile node with station count mechanism updates the number of stations it found in a BSS in each short sniffing period (e.g., 20 ms) of a discrete scan scheme. Because the mobile node concerns only the stations that locate in its coverage area and transmit at least one frame in a short sniffing period, much less cache memory is

needed to keep MAC addresses of discovered stations, and a simple algorithm that collects the frames of various senders to a receiver is sufficient to identify an active station in the coverage. Furthermore, a mobile node starts to sense a BSS before it enters the coverage; therefore the evaluation of a BSS is available with discrete scan once its AP is discovered by the mobile node.

As illustrated in Figure 3.5(a), the nearer is the mobile node to an AP, the larger overlapped coverage the mobile node can listen, and consequently the mobile node can discover more active stations in a sniffing period. Unfortunately, the distance between a mobile node and an AP is not a unique factor that affects the station counts. As illustrated in Figure 3.5(b), the total number of stations in a BSS is another major factor that affects the station counts in a sniffing period. Due to the imbalance of number of stations in two BSSs, the farer AP may result in larger station count than a nearer one



(a)

(b)

Figure 3.5 The station counts in two BSSs

- (a) The larger station count maps to the closer AP.
- (b) Imbalanced number of stations of two BSSs may reverse the results

The third factor that affects the station counts in a sniffing period is the length of a sniffing period. The longer sniffing time results in larger station count until the mobile node discovers all of the stations in its coverage. On the other hand, if the sniffing period is as short as a slot time in DCF wireless environment, the station count is reduced to the probability for one of the stations in the mobile node's coverage to get a successful transmission in the given timeslot. Since all of the stations in a BSS share a media of limited bandwidth, a BSS of fewer stations has higher probability to transmit a frame in an arbitrary timeslot because of less collisions and smaller contention windows. Hence, a short sniffing period benefits a BSS containing fewer stations.

The length of sniffing periods of discrete scan is properly set to either balance or alleviate the affect of the station counts brought by imbalance of number of stations in various BSSs. There are two major concerns on setting the length of a sniffing period: first, a long sniffing period may eventually eliminate the affect brought by imbalanced number of stations. Second, a short sniffing time results in insufficient station counts to have creditable selection among the candidate APs.

In the next section, we will elaborate the way to determine an appropriate length of a sniffing period. The model proposed by G. Bianchi [10] is employed for the analysis of DCF wireless LAN environment. Furthermore, with ns-2 simulation results, those for EDCA wireless environment are demonstrated in Chapter 4.

3.3.2 Setting of a Sniffing Period

As described in previous section and illustrated in Figure 3.5(a) (b), two factors control the station count of a BSS, distances between the mobile node and the AP and

number of stations in the BSS. Intuitively, we have

$$E(s) \approx r \cdot A' / A \tag{1}$$

where, E(s) denotes the expected value for s, and s denotes the station count of a BSS in a sniffing period that is represented by T_{sniff} .

> r is a new term called as "number of transient stations", that is defined as the number of stations which at least send one frame in given sniffing time (T_{sniff}) in a BSS.

> A' denotes the overlapped coverage of a BSS and the mobile node, and A denotes the coverage area of BSS.

Let τ denote the probability that a station transmits in an arbitrary slot time, and p denotes the probability that a transmitted packet encounters a collision. By [10], in a BSS of DCF wireless environment, there are

$$\tau = \frac{2(1-2p)}{(1-2p)(W+1) + pW(1-(2p)^m)}$$
(2)
$$p = 1 - (1-\tau)^{n-1}$$
(3)

 $p = 1 - (1 - \tau)$

W is the minimal contention window in DCF.

where, *n* is the number of stations in a BSS

m is the integer such that $2^m W$ presents the maximal contention window in DCF.

For a given n, τ and p are derived from (2) and (3). Thus, the average number of frames transmitted within one timeslot would be τ *n.

Let P_{tr} denote the probability that at least one frame is sent in a considered timeslot. There are *n* stations and each station transmits a frame with probability τ , therefore,

$$P_{tr} = 1 - (1 - \tau)^n \tag{4}$$

Let P_s denote the probability of a successful transmission in a considered timeslot, i.e., the probability that exactly one station transmits on the channel, under the condition of the fact that at least one station transmits. It has,

$$P_{s} = \frac{n\tau(1-\tau)^{n-1}}{P_{tr}} = \frac{n\tau(1-\tau)^{n-1}}{1-(1-\tau)^{n}}$$
(5)

Let σ be a slot time in the condition of idle media. T_s denotes the length of a slot time in the condition of a successful transmission and T_c denotes the length of a slot time in the condition of a transmission with collision, as defined in Figure 3.6.

$$T_{s} = PHYhdr + MAChdr + E[P] + SIFS + \delta + ACK + DIFS + \delta$$
(6)

$$T_c = PHYhdr + MAChdr + E[P^*] + DIFS + \delta$$
(7)

where, δ stands for the propagation delay, E[P] is the average length of packet payload and $E[P^*]$ is the length of the longest packet payload involved in a collision.



Figure 3.6 Definition of T_s and T_c , slot time of transmission with success and collision.

The average length of a slot time, denoted by T_{av} , can be readily obtained by considering that, with probability $(1-P_{tr})$, the slot is null of transmission; with probability P_{tr} P_s , the slot contains a successful transmission, and with probability $P_{tr}(1-P_{tr})$, the slot contains a collision. Hence,

$$T_{av} = (1 - P_{tr})\sigma + P_{tr}P_sT_s + P_{tr}(1 - P_s)T_c$$
(8)

Let *m* denote the total number of frames sent by *n* stations within T_{sniff} . It is estimated in average as,

$$m = n\tau * T_{sniff} / T_{av}$$
⁽⁹⁾

Assume that *n* stations transmit *m* frames randomly. The number of transient stations, *r*, is the number of stations that send at the least one of *m* frames in T_{sniff} . Thus, *r* is derived as follows:

$$r = \frac{\sum_{k=1}^{n} k \cdot P(n,k) \cdot S(m,k)}{\sum_{k=1}^{n} P(n,k) \cdot S(m,k)}$$
(10)

where, S(m,n) denotes the Stirling number of the second kind, as,

$$S(m,n) = \frac{1}{n!} \sum_{k=0}^{n} (-1)^{k} {\binom{n}{n-k}} (n-k)^{m}$$
(11)

and P(n,k) denotes the permutation number as,

$$P(n,k) = \frac{n!}{(n-k)!}$$
(12)

Note that m in (9) is not an integer but m in (10) and (11) is restricted to an integer. Interpolation will be used in the numerical approach below.

With equations (2)~(12) and parameters listed in Table 3.1 for an IEEE 802.11b wireless LAN, the relations of transient number of station(r) v.s. number of station(n) of various T_{sniff} values are demonstrated in Figure 3.7.

Parameters	Values
PLCP preamble & header	192.0 <i>µs</i>
MAC header	20.4 µs
Slot time	20 µs
SIFS	10 µs

DIFS	50 µs
ACK	10.2 μ s + PLCP preamble & header
CW_{min}	32
CW _{max}	1023
Maximal Packet payload	2312 bytes
Channel bit rate	11 Mbps

Table 3.1 Parameters for IEEE 802.11b wireless LAN



Figure 3.7 Number of transient stations (r) v.s. number of stations (n) for different T_{sniff} values.

By reviewing equation (1), a station count (s) linearly depends on overlapped area (A') and transient number of station(r) with respect to a T_{sniff} . Since it is desired

that a station count to indicates overlapped area (A') rather than being disturbed by the imbalance of numbers of stations in different BSSs, an appropriate sniffing period (T_{sniff}) should be set so that the number of transient stations (r) behaves insensitively to variation of the number of stations (n).

Figure 3.7 shows the fact that a shorter sniffing period (T_{sniff}) responds to a number of transient stations (r) less insensitive to variation of number of stations (n). However, as described in the previous section, a sniffing period should be long enough to make station counts big enough for a creditable selection. In this thesis, we recommend that in a BSS of fewer stations (such as n = 6), number of transient stations (r) should be 80 % of number of stations (n), i.e., r > 4.8 as n = 6. From the curves in Figure 3.7, a sniffing period T_{sniff} is eventually chosen as 20 ms because the curve of $T_{sniff} = 20$ ms shows that its transient stations equal to 5 with n = 6. As n = 24, the number of stations increases by 4 times, the number of transient stations (r) is approximately equal to 10, increased by only 2 times. The numerical result shows that the variation by setting a sniffing period as 20 ms. Although there is no proper sniffing period to make number of transient stations independent of number of station, however, the effect brought by imbalance of number of stations has been significantly mitigated by 50%

3.3.3 Performance of Station Count Mechanism

E(s), the expected value of the station count with respect to a given T_{sniff} , is approximately equal to r^*A'/A as described in (1), while collisions within overlapped area, A', is neglected. In this section, the collisions are taken into consideration for a more accurate formula to estimate E(s).

The average number of stations lies in the overlapped coverage that is estimated by area proportion, so that $n' = n^*A'/A$. Let *m*' denote the total frames sent by either one of *n*' stations (no collision with any frame sent by one of *n*' stations) in T_{sniff} , then *m*' is estimated in (13), where τ is given in (2) and *m* is given in (9)

$$m' = n' \tau (1 - \tau)^{n' - 1} \cdot T_{sniff} / T_{av} = m \cdot (1 - \tau)^{n' - 1} \cdot n' / n$$
(13)

Same as the process to estimate the number of transient stations, assuming that n' stations transmit m' frames randomly and let μ denote the reformed station count, the number of stations that sends at least one of m' frames in T_{sniff} . Thus, μ can be derived using (14) below.

$$\mu = \frac{\sum_{k=1}^{n} k \cdot P(n',k) \cdot S(m',k)}{\sum_{k=1}^{n} P(n',k) \cdot S(m',k)}$$
(14)
where, $S(m,n)$ denotes the Stirling number of the second kind that is
defined in (11), and $P(n,k)$ denotes permutation number that
is defined in (12)

Note that m' in (13) may not be an integer and m' in (14) is restricted to an integer. Interpolation is used for numerical approach below.

Let S_i denote the random variable of station count in a sniffing period T_{sniff} for i^{th} channel. It is reasonably assumed that

- i S_i features Poisson distribution.
- ii μ_i , derived in (14), an estimation of reformed station count for i^{th} channel, should be the average of random variable S_i , i.e., $\mu_i = E(s_i)$.

The scenario to display the performance of station count mechanism is illustrated in Figure 3.8. It assumes that a mobile node with discrete scan scheme roams in the overlapped coverage of two next-AP candidates. The current AP is not shown in Figure 3.8, for its location will be irrelevant to performance of the station count mechanism. The mobile node selects one of two next-AP candidates, AP₁ and AP₂, according to their station counts derived in the last two sniffing periods of the discrete scan. Let *h* denote the hit ratio, the probability that the mobile node selects the nearest AP with the indication of larger station count as its next AP. Therefore, the mathematical definition for *h* in the case of two candidates is given in (15).

$$h = p(s_1 > s_2 \mid d_1 < d_2) + \frac{1}{2}p(s_1 = s_2 \mid d_1 < d_2)$$
(15)

where, p(...|...) denotes the conditioned probability.

 s_i denotes the station count in T_{sniff} of *ith* channel.

 d_i denotes the distance between the mobile node and the AP in channel *i* (denoted by AP_i).



Figure 3.8 Scenario for performance estimation.

The average station count in each channel, μ_1 and μ_2 , can be estimated by (14) with given n_1 , n_2 , d_1 , d_2 , T_{sniff} as wells as parameters for DCF wireless environment (listed in Table 3.1). Since S₁ and S₂ are in Poisson distribution of mean values $E(s_1) = \mu_1$ and $E(s_2) = \mu_2$, the hit ratio *h*, defined in (15) is estimated by,

$$h = \sum_{k=1}^{\infty} p(s_1 = k) \cdot [p(s_2 < k) + \frac{1}{2} p(s_2 = k)]$$

$$=e^{-(\mu_1+\mu_2)}\sum_{k=1}^{\infty}\left[\frac{\mu_1^k}{k!}\left(\sum_{l=0}^{k-1}\frac{\mu_2^l}{l!}+\frac{\mu_2^k}{2k!}\right)\right]$$
(16)

Hit ratios are demonstrated for the conditions that the mobile node locates at the place $d_1 = 0.5R$ and d_2 varies from 0.5R to 1.2R, where R denotes the radius of coverage of BSSs and the mobile node. T_{sniff} is set as 20 ms. In the first case, it demonstrates how the distances between the mobile node and APs affect the hit ratio by setting the same number of stations in BSSs to ignore effects of the imbalance of numbers of stations in BSSs. In Figure 3.9, the curves are given for hit ratios (*h*) with respect to various distances from AP₂ (d_2) in cases of $n_1 = n_2 = 6$, 12, 18 and 24.



Figure 3.9 Hit ratios v.s. distances in balanced number of stations.

Figure 3.9 shows that the hit ratios increase almost linearly with the increase of distance between the mobile node and the competing AP (d_2) , while the distance from the mobile node to the selected AP (d_1) is set fixed and the effect of imbalanced number of stations are ignored, i.e., $n_1 = n_2$. From observation of Figure 3.9, it indicates that the number of stations, if they are evenly distributed, will be almost

independent to the hit ratios.

In the second case, we investigate the affect of imbalanced number of stations on the hit ratios. The mobile node keeps its location at the place $d_1 = 0.5R$ and $d_2 = 0.9R$. Four cases of variation of number of stations in BSS₁, $n_1 = 6$, 8, 10 and 12 are individually input. For each case, the number of stations in BSS₂, n_2 , varies from 0.6 to 2 times of n_1 and the corresponding hit ratios are computed and presented in curves, as shown in Figure 3.10.



Figure 3.10 Variation of hit ratios with imbalance of number of stations.

Observe the curves in Figure 3.9, the imbalance of number of stations do affect the hit ratios. In the neutral number of stations, i.e., $n_2 = n_1$, the hit ratios for all of the four cases are about 0.68. With the increase of imbalance, i.e., increasing n_2 / n_1 , the hit ratios decrease with moderate slopes. As to the most imbalanced number of stations, i.e., case of $n_2 = 2n_1$, the hit ratios drop to about 0.55~0.6. The hit ratio decreases by less than 20% due to the number of stations of competing BSS increases up to double. In addition, it is more sensitive to the disturbance of imbalance in number of stations in case of fewer stations in the selected BSS (BBS₁). In Figure 3.10, in the case that most of stations are in selected BSS (i.e., $n_1 = 12$), the hit ratio drops to about 10% (from 0.68 to 0.6) with the increase of imbalance (n_2/n_1) from 1.0 to 2.0. On the other hand, the hit ratio drops to about 20% (from 0.68 to 0.55) in case that there are fewest stations in the selected BSS ($n_1 = 6$).

3.3.4 Select Best Next AP Instead of the Nearest AP

The traditional handoff scheme that determines the next AP to handoff to by the RSS (receive signal strength) received from the responding AP in the probe phase of a layer-2 handoff. The strongest RSS is recognized as an indicator of the nearest AP, because shorter distance is subject to stronger signal strength. Same as RSS, the mechanisms presented in section 3.3.1 help a roaming node select the nearest AP by the link-layer information obtained in discrete scans. However, the nearest AP may not be the best next AP to handoff to. The available bandwidth that the next BSS can offer is one of the most concerned items especially for a mobile node in real time application, such as a WiFi VoIP connection. Besides, an AP that the mobile node is approaching to may be a better selection of the next AP than those from which the mobile node is moving away, even the latter ones are detected with stronger RSS.

Discrete scan schemes may be used to detect characteristics of a BSS by the collection of frames transmitted in a BSS in advance to a handoff. With information from sniffed frames, a mobile node may select the best next AP rather than the possible nearest AP with the indication of RSS in a traditional handoff scheme.

The available bandwidth and access delay of a BSS can be detected by the average NAV values in the frames collected in discrete scans. The relations between

the NAV values and available bandwidth as well as access delay in a BSS are inferred to be linearly dependent in [11] with both mathematical analysis and simulation. A mobile node with discrete scan will be able to estimate the available bandwidths and access delays of the next-BSS candidates by NAVs collected in its sniffing periods and determine the best next AP when a handoff is triggered.

Furthermore, the trends of station counts along certain sniffing periods for a BSS may be used to indicate whether the sniffing node is approaching to an AP or not. For the AP that a mobile node is approaching to, the station counts shall be in an increasing trend because the mobile node can listen to more stations when it is closing to the AP. To select an approaching AP rather than the nearest AP may alleviate the frequent handoffs in the hot spot where the cell of a BSS may be relatively small.

The attributes that assist a mobile node to determine its next AP in handoffs may be integrated as an objective function with appropriate weights assigned to all of the indicators. With combinational considerations on receive signal strength, distances, approaches, as well as available bandwidth, the objective function for a mobile node to evaluate i^{th} BSS can be written as,

$$F_i = w_1(RSS)_i + w_2(STA_count)_i + w_3(\Delta STA_count)_i + w_4(NAV)_i$$
(17)

3.4 Impact on Service QoS

The most concerns on the discrete scan schemes may attribute to its impact on the service QoS in a pre-handoff period, since part of service time is taken out to sniff on other channels. During the sniffing periods, the service of working channel will be interrupted and this causes a certain level of access delay. Furthermore, because of absence from working channel, the frames that the current AP sends to the mobile node during a sniffing period will be lost. However, as described in chapter 1, the applications that require seamless handoffs are mostly of relative low bit rate as compared with that a wireless NIC can offer. Therefore, with proper design, a mobile node may use a large portion of idle time of wireless NIC for discrete scan; thus, to minimize the impacts caused by discrete scan. In this section, a VoIP connection which runs with bidirectional 64 kbps constant bit rate in an IEEE 802.11b wireless LAN of 11Mbps are taken as an example for the discussion on the impact on service QoS brought by discrete scan scheme.

Besides a sniffing period of 20 ms, the channel switch time of a wireless NIC contributes to disruption of service from working channel as a major part. In this thesis, the channel switch time is assumed as 5 ms as in [1]. To complete a sniffing period in discrete scan scheme, it needs to do twice for switching channels, one for leaving from the working channel and the other for returning back. Therefore, it brings an absence period of at least 30 ms from the working channel to execute once of discrete scan.

A disruption of 50 ms may be the upper bound for a seamless handoff to tolerate. The principle of the discrete scan is to decompose time for the probe phase of a layer-2 handoff into a pre-handoff period, such that no more than 50 ms disruptions are induced during pre-handoff and handoff procedures.

Nowadays, the QoS issue for real time applications in IEEE 802.11 wireless LAN is still widely discussed. In most of infrastructure wireless LAN environment, downlink traffics from AP to all the stations is much heavier than the uplink traffic from a single station to the AP. However, the transmission opportunity of AP is same as that of a single station, thus the downlink flow shares the bandwidth with those uplink flows from all stations. In other words, the shared bandwidth of the downlink transmission from AP to arbitrary one of stations will be only 1/n of the bandwidth of

the uplink transmission to AP, where *n* denotes number of stations in a BSS. Because of the asymmetric nature between uplink and downlink transmission, a real time application with symmetric bidirectional transmission may suffer from severely degraded QoS of downlink flow when the number of stations in a BSS increases.

To address the QoS problems caused by asymmetric transmission opportunity for a specific station with symmetric bidirectional traffics, piggyback schemes [12] in IEEE 802.11 standard suppose to forward the downlink traffic as an attachment to a positive ACK frames from AP to stations. With piggyback schemes, the QoS issue of asymmetric transmission opportunity is therefore eliminated. The QoS provisioning to symmetric bidirectional service then is considered as the QoS to the uplink flow of the concerned station.



Figure 3.11 Total disruptions induced by a discrete scan.

As illustrated in Figure 3.11, the total disruption caused by discrete scans should be computed from the transmission of the last frames before wireless NIC switches to a sniffing channel until the transmission of the first frame after wireless NIC returns back to its original working channel. The duration for a wireless NIC absent from working channel is 30 ms for each sniffing period and the total disruption has to be kept less than 50 ms, therefore, the two sentinel frames, defined as the last frame before channel switch and the first frame after the return, have to be transmitted within the periods of 20 ms in total.

The rest of 20 ms for sentinel frames is divided into division A and B (shown in Figure 3.11). A mobile node converses a disruption caused by a sniffing period less than 50 ms, if it transmits the first sentinel frame in division A and the second sentinel frame in division B. In the case the first sentinel frame fails to be transmitted with the division A period, the piece of sniffing period should be slipped until next cycle. As the first sentinel frame has been sent, the total disruption brought by the sniffing period is equal to 50 ms + (time to the 2^{nd} sentinel frame – division B) – time to the 1^{st} sentinel frame. To keep the disruption less than 50 ms in all case, a mobile node should promise to send the second sentinel frame by the end of division B.

In the simplest design, each sentinel frame for a sniffing period evenly shares 10 ms as its maximal allowable time for the mobile node to transmit a sentinel frame successfully. Thus we have division A = division B = 10 ms. Therefore, the capability that a mobile node guarantees to send one frame successful in 10 ms decides whether the disruptions brought by a discrete scan can be conserved within 50 ms. In this section, the analytical model of DCF wireless environment presented in section 3.3.1 is employed to estimate the probability for a mobile node in IEEE 802.11b wireless LAN to transmit one frame successfully within 10 ms.

With known transmitting probability in a slot time, τ , given in (2) and the average length of a slot time, T_{av} , given in (8), the average number of frames that a mobile node could transmit within 10 ms can be derived as,

$$m'' = \tau * 10ms / T_{av}$$
 (18)

Taking the probability that a transmitted packet encounters a collision, p, given in (5), the probability that at least one of m " frames transmitted without a collision is estimated by,

$$p_{\text{sentinel}} = 1 - p^{m''} \tag{19}$$

Numerical results of (19) with the parameters listed in Table 3.1 for BSSs of 11 Mbps as well as 54 Mbps are given in Figure 3.12. When number of stations increases up to 6, the probability for a station to transmit a frame without collisions within 10 ms drops to lower than 80% in an IEEE 802.11b 11Mbps wireless LAN, therefore, more than 20% of sniffing periods will be slipped. It is inferred about 20% of disruptions brought by sniffing periods to be longer than 50 ms when the BSS supports more than 6 stations simultaneously. Figure 3.12 shows that even bit rate of the BSS increases up to 54 Mbps, stations that the BSS can support based on the same criteria increases to only 12. Note that, as shown in Figure 3.9 and 3.10, the station count mechanism proposed in section 3.1.1 assumes a number of stations greater 6 to promise creditable selections. It consequently concludes that the discrete scan with station count mechanism proposed in 3.1.1 will not work together very well in the DCF environments. However, instead of DCF, an IEEE 802.11e EDCA wireless environment grants VoIP traffic with high priority of transmission and promises it to send one frame within 10 ms at a much higher probability even with more than 10 stations in the BSS.

How discrete scan and station count mechanism proposed in this thesis can work together in an IEEE 802.11e ECDA wireless environment will be elaborated in Chapter 4 with results of ns-2 simulation. To set a proper length for a sniff period, the performance to hit the desired nearest AP as well as probabilities that the disruptions for sniffing periods converge within 50 ms will be demonstrated.



Figure 3.12 Probability to transmit a frame in 10 ms v.s. number of stations in a BSS.



Chapter 4 Performance Evaluation

In this chapter, we evaluate the performance for discrete scan schemes in IEEE 802.11e EDCA wireless LAN environment by means of simulations. In section 4.1, we introduce the simulation environment and concerned settings in the simulations. The discrete scan scheme for collection of next-AP candidates for an IEEE 802.11 DCF wireless LAN environment described in section 3.1 and 3.2 is still valid to be applied in an IEEE 802.11e EDCA wireless environment; therefore, there is no further supplement in this chapter. In section 4.2, it applies station count mechanism proposed in section 3.3 for the selection of the nearest AP as the next AP to handoff to in IEEE 802.11e EDCA wireless LAN environment. Instead of using the analytical model for IEEE 802.11b DCF wireless LAN, it utilizes ns-2 simulator to evaluate the performance of the prediction mechanism as those presented in section 3.3.3. In section 4.3, we discuss the major concern, the impact of service QoS, while a mobile node applies discrete scan scheme in EDCA wireless LAN. Attributing to the high transmission priority of VoIP traffic, the impact on QoS caused by discrete scan schemes in EDCA environment is shown within the tolerable limits according to the simulation results.

4.1 Simulation Environment

To demonstrate the performance of discrete scan schemes with station count mechanism in IEEE 802.11e EDCA environment, we use NS-2 (version 2.28) tool [13] with 802.11e tkn EDCA module [14]. To simplify the simulation, we neglect

high-level management functionality such as beacon frames, association and authentication frame exchanges. In all scenarios, the network topology of the simulations is shown in Figure 4.1. Each of wireless stations either runs bidirectional VoIP traffic or TCP traffic with its corresponding wired station playing as either a VoIP device or FTP server. VoIP traffic, with format of G.711 codec, 160 bytes payload and 20ms intervals are used as real-time traffic, and FTP traffic of 1,500 bytes payload are used to simulate the best effort traffic. Table 4.1 shows the default parameter values of EDCA in the simulations. Besides, 10% of wireless packet error rate is set to simulate the transmission in the real environment. Furthermore, for the sake of distributed arrival time of VoIP packets, each of VoIP traffic is initiated at a randomly selected time from 5^{th} to 6^{th} second. The observed scan periods for each channel are taken from 10^{th} second to 20^{th} second.



Figure 4.1 Simulation configuration

With various random seeds for each run, uniformly distributed random function, which is built-in in ns-2 programs, generates the locations of the mobile nodes in both BSSs. The normal combination of traffics is assumed that 70% of nodes run in TCP traffic and 30% of nodes play as VoIP phones. However, BSSs of all TCP nodes and all VoIP nodes are running in order to discuss the affects by various traffics in a BSS.

AC	PF	AIFS	CW_MIN	CW_MAX	TXOP Limit (s)
voice	2	2	7	15	0.003008
best effort	2	3	31	1023	0

Table 4.1 Default values of parameters for original EDCA.

4.2 Station Count Mechanism in EDCA Environment

4.2.1 Setting of a Sniffing Period in EDCA Environment

As discussed in section 3.3.2, by properly selecting a sniffing time may eliminate affects on the hit ratio in selecting the nearest AP caused by imbalance of number of stations between the next-BSS candidates. The number of transient stations (denoted by r) is defined as the average number of stations that transmit at least once within a sniffing period (denoted by T_{sniff}) in a BSS. A proper sniffing time shall meet two conditions: First, making the number of transient stations varies as insensitively as possible with the variation of number of station in a BSS. Usually the less the sniff time is, the better the attribute will be. Second, the sniff time should be long enough for a mobile node to discover the next-AP candidates as well as to collect enough frames to infer a creditable selection result.

In Figure 4.2, it shows the curves for number of transient stations v.s. various numbers of stations for several given values of sniffing periods in an EDCA BSS of 70% TCP and 30% VoIP nodes. The curves in Figure 4.2 show that number of transient stations is fairly insensitive to number of stations as the number of stations in a BSS is large (e.g. n > 12). When the number of stations in a BSS is small (e.g. n < 6), the number of transient stations increases at a rate about a half of increasing rate of number of stations. The selection results are still disturbed by the imbalance of number of stations, especially when number of stations is small in both EDCA and DCF wireless environment. However, the effect is reduced to at least half of the difference of number of stations by setting a sniff period to $15 \sim 20$ ms.



Figure 4.2 Number of transient stations v.s. number of stations in EDCA of 70% TCP and 30% VoIP nodes.

Figure 4.3 shows the curves with the same condition as those in Figure 4.2

except that EDCA BSS features 100% of TCP nodes. With mostly the same trends as that in Figure 4.2, the curves in Figure 4.3 shows less steep slope and apparently lower values of number of transient stations than those in Figure 4.2. It is inferred that most VoIP nodes are discovered and contribute to number of transient stations because of the attributes of high transmission priority and short packet length of VoIP traffic. Consequently, the station count mechanism tends to select a BSS with more VoIP nodes as the next AP to handoff to if the other control factors, such as distances and number of stations are neutral between BSSs.



Figure 4.3 Number of transient stations v.s. number of stations in EDCA with 100% TCP nodes.

4.2.2 Performance of Station Count Mechanism in EDCA

Environment

Figure 4.4 shows that the hit ratios with respect to various distances ($d_2 = 0.5R \sim 1.1R$) from the competing APs. The selected AP locates at a fixed distance ($d_1 = 0.5R$) from the mobile node. The numbers of stations in both BSSs are set to the same set of numbers ($n_1 = n_2 = 6$, 12, 18, 24) to minimize the effects caused by imbalanced number of stations. Both next-BSS candidates work as EDCA wireless LANs consisting of 70% TCP and 30% VoIP nodes.



Figure 4.4 Hit ratios v.s. distances in balanced number of stations in

EDCA WLAN of 30% VoIP nodes and 70% TCP nodes.

With the same situation as those in DCF environment (shown in Figure 3.9), the hit ratios increase linearly with the increase of the distance between the mobile node and the competing APs, d_2 , while d_1 is fixed and numbers of stations are evenly distributed. It is inferred that the distances between a mobile node and next-AP

candidates affect the hit ratios to select the nearest AP regardless the type of wireless LANs, with the assumption of evenly distributed stations. Consequently, it concludes that the station count mechanism works under EDCA as well as under DCF.

Figure 4.5 shows the hit ratio that is same as condition in Figure 4.4 except that the EDCA BSS consists of 100% TCP nodes. Compared with that in Figure 4.4 and Figure 3.9, it is inferred again that the distance dominates the hit ratios linearly. However, the deviation among curves of numbers of stations (n_1 and n_2) is apparently larger than that in Figure 4.5. As described in previous section, a mobile node will discover most of the VoIP nodes in its coverage; therefore, these VoIP nodes contribute to the station count. The curves in Figure 4.5 display a trend much similar to those in Figure 3.9 because EDCA is reduced to DCF if only best effort access category (AC) exist in an EDCA BSS.



Figure 4.5 Hit ratios v.s. distances in balanced number of stations in EDCA WLAN of 100% TCP traffic.

Furthermore, with existence of VoIP nodes in EDCA, small portion (e.g. 30%)

of VoIP nodes dominates the selection of the next AP. The station count mechanism may select a BSS of more VoIP nodes rather than the nearest one.

Figure 4.6 shows the effects on hit ratios caused imbalance number of stations in an EDCA BSS of 70% TCP nodes and 30% VoIP nodes. Same as those in Figure 3.10, the mobile node locates at the place $d_1 = 0.5R$ and $d_2 = 0.9R$, the number of stations of BSS₁, $n_1 = 6$, 8, 10 and 12, and the imbalance of number of stations of BSS₂, n_2/n_1 varying from 0.6 to 2.



Figure 4.6 Hit ratios v.s. imbalanced number of stations in EDCA of 30% VoIP nodes and 70% TCP nodes.

Figure 4.7 shows the hit ratios of the same settings in Figure 4.6 except that the EDCA has only TCP nodes. Figure 4.6 and Figure 4.7 show the interference of imbalanced number of stations in the next EDCA BSSs candidates on the hit ratios of station count mechanism are of the same averages and trends, except that the curves in Figure 4.6 are of larger deviation from their average values. As interpreted above,

the existence of VoIP nodes contribute the deviation to the curves in Figure 4.6, because few VoIP nodes dominate the selection results.



Figure 4.7 Hit ratios v.s. imbalance of number of stations in EDCA of 100% TCP traffic.

With the discussion above, the existence of VoIP nodes may contribute a negative factor for station count mechanism to select the nearest AP. Generally speaking, VoIP packets are generated in an interval of 20 ms and transmitted within bounded jitter to meet real-time requirement. Since a sniffing period is suggested to be as long as 20 ms, it implies that all VoIP nodes transmit a frame each sniffing period. Therefore, the number of VoIP nodes is included in the number of transient stations in an EDCA BSS, then, it causes station count mechanism intending to select a BSS of more VoIP nodes rather than the nearest one.

It is still not yet to comment as a good or bad feature for station count mechanism to select an AP of more VoIP nodes in its BSS rather than the nearest one.

However, the feature can be designed as an option because of the simplicity to filter out the frames from VoIP nodes by the specific characteristic of its packets. As the frames for VoIP traffic are ignored in the sniffing periods, the performance of the station count mechanism in EDCA environment reforms to be similar to that in DCF and with less deviation from the mean values, as shown in Figure 4.5 and Figure 4.7.

4.3 Impact on Service QoS in EDCA Environment

Referring to the discussion in section 3.4 and the scheme shown in Figure 3.11, to fulfill the seamless requirement and protect services for a disruption of more than 50ms, a VoIP node with discrete scan scheme shall ensure to transmit one frame successfully within 10 ms before and after a sniffing period. In an EDCA wireless LAN, VoIP traffic is granted with the highest transmission priority, therefore, the contentions for a VoIP flow in EDCA wireless LAN to get medium are much less intense than those in DCF one. In Figure 4.8, it shows the probabilities for a VoIP node to send at least one frame in 10 ms with respect to the number of stations in an EDCA wireless LAN with 11Mbps rates. Three cases for further discussion are assumed: first, 10% of transmission error in the wireless environment and 100% of stations in a BSS are VoIP nodes; second, 10% of transmission error in the wireless environment and 100% of VoIP nodes for all stations in a BSS.

The curves in Figure 4.8 display that the probabilities for a VoIP node to send a frame in 10 ms in the EDCA environment are more dependent on the air transmission error rather than on the number of stations in its BSS. In worst condition such that

more than 20 VoIP running simultaneously in an EDCA BSS with 10% transmission error, the success probability to send a frame in 10 ms is still kept at around 85%. Therefore, the disruption caused by a sniffing period for discrete scan will be maintained within 50 ms with a probability more than 85%. As compared with those for DCF environment, it concludes that the discrete scan schemes proposed in this thesis are much applicable for a VoIP node in EDCA wireless environment than in DCF one.



Figure 4.8 Probability to send a frame in 10 ms v.s. number of stations in 11 Mbps EDCA wireless LAN.

Chapter 5 Conclusion and Future Works

In this thesis, the concept of "discrete scan" is newly proposed. Based on the discrete scan scheme, the next AP in a handoff can be discovered prior to the occurrence of handoffs. With the survey of fast handoff schemes based on the prediction of next APs, a series of handoff schemes, including context caching, proactive key distribution, buffering and forwarding, cross-layer fast handoff design, are suitable to cooperate with discrete scan scheme to fasten layer-2 and layer-3 handoff to achieve the goal of seamless handoffs.

The principle of discrete scan lies on the decomposition of the passive scan procedure in a traditional layer-2 handoff into discrete pieces of sniffing periods prior to the occurrence of handoff. A mobile node utilizes the possible idle time of its NIC as the sniffing periods so that the disruption caused by a sniffing activity is controlled less than 50 ms to fulfill the minimal requirement of seamless handoff features.

With the application of discrete scan scheme, a mobile node can further select an appropriate next AP among the next-AP candidates with the help of information extracted from MAC headers in those frames collected in sniffing periods. Several mechanisms to select the next AP with desired features are proposed in this thesis, such as the ratio of stations located within coverage of the mobile node to that in the BSS and the station count discovered in the last sniffing period before a handoff is initiated to select the nearest AP as the next AP. Besides, a scheme to identify an approached AP and an indicator to predict the available bandwidth of concerned BSS are briefly discussed in this thesis.

With the benefits of simplicity and nearly real-time characteristic, the station

count mechanism is elaborated and evaluated in this thesis. The setting of a sniffing period for discrete scan scheme, performance for the mechanism to hit the nearest AP as well as the impact on service QoS caused by the absence from working channel in sniffing periods for discrete scan scheme in a DCF wireless environment are demonstrated with both analytical model and numerical results. For further verification, those for EDCA wireless environment are analyzed with results generated in ns-2 simulations.

The numerical as well as the simulation results show that discrete scan with station count mechanism provides considerable high hit ratio on the selection of the nearest AP among the next-AP candidates discovered by a mobile node. Although the imbalance of number of stations in the next-AP candidates is still an inevitable factor to affect the selection accuracy for the nearest AP, a proper length of the sniffing periods can significantly reduce the effects.

The nature that an AP, responsible for all downlink traffic, has the same transmission opportunity as an arbitrary station, taking care only its own uplink traffic, in a BSS may causes insufficient downlink bandwidth for a symmetric bidirectional connection such as a VoIP application. The downlink problem for VoIP traffic is supposed to be resolved with the piggyback scheme reported in 802.11e standard, because it is an inherent QoS problem instead of being caused by discrete scan scheme. However, the impact on the service QoS caused by discrete scan scheme is discussed in view of uplink flow for VoIP traffic. Simulation results show the induced disruptions being maintained within a tolerable level in the EDCA environment for all cases of reasonable number of stations in a BSS; However, numerical analysis shows a mobile node with discrete scan scheme in DCF wireless LAN can sustain acceptable QoS when less than six stations are in a BSS.

In the future, we will further improve the scheme by investigating a scenario that

a mobile VoIP node with discrete scan scheme works in EDCA environment. The observation will focus on the QoS degradation induced by discrete scan scheme. Besides, an integrated selection mechanism with sniffing function and an algorithm to identify a station via reception of its transmitting frame will be developed. Finally, we will plan to implement a complete set of discrete scan as well as selection mechanism into a real WiFi phone. The WiFi phone shall be proved with experiments the capability of fast layer-2 handoff because latency of probe phase has been eliminated. The performance to select a desired next AP shall be verified with experiments. Eventually, the strategies addressed for fast handoff in the related works, such as context caching, buffering-and-forwarding and cross-layer fast handoff design will be implemented into the APs and FAs to cooperate with a mobile node with discrete scan scheme, so that a complete wireless environment can be configured to support seamless inter-domain handoff for a next generation WiFi streaming services.



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