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The Study Adaptive Channel Estimation in

MIMO-OFDM based LTE Modem

研究生:黄貴英

指導教授:許騰尹 教授

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應用於 LTE 收發器之多輸入多輸出正交分頻多工系統的

可適性通道估測設計

The Study of Adaptive Channel Estimation in MIMO-OFDM based LTE Modem

研究生:黃貴英Student: Guei-Ying Huang指導教授:許騰尹Advisor: Terng-Yin Hsu

國立交通大學



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摘要

本論文研製了一個可應用於 3GPP Long-Term Evolution(LTE) 無線網路標準 的可適性通道估計演算法,此可適性通道估測演算法可對抗時變之頻率選擇性快 速衰減通道。LTE 無線網路標準為設計時的參考平台,2對2的多重輸入多重輸 出正交分頻多工系統被納入考慮。

本論文提出的通道估測演算法包含兩個步驟。第一個步驟利用 PDCCH 初步估 測通道後,再經由本論文所提出的可適性通道估計演算法做處理,以估計出包含 雜訊的時變通道,再由頻域上的一階低通濾波器濾去部分的雜訊,以降低演算法 對訊噪比的需求。

根據模擬的結果,此可適性等化器在2對2的多輸入多輸出系統下,可在傑 克斯模型 120km/hr、訊噪比 33dB 的條件下可達到與 One-shot 通道估測演算法 相同的效能,且與 One-shot 通道估測演算法比較,乘法數比 One-shot 通道估測 演算法減少 88%,且記憶體的容量為 One-shot 通道估測演算法的 1/14。



Abstract

An adaptive channel estimation algorithm applied to 3GPP Long-Term Evolution (LTE) protocol under time-variant frequency-selective fast fading channels is proposed in this thesis. LTE wireless internet protocol is the design reference and 2x2 multiple inputs multiple outputs (MIMO) Orthogonal Frequency Division Multiplexing (OFDM) are considered.

There are two steps in the presented method. Firstly, the system estimates the channels by PDCCH. Secondly, the proposed adaptive channel estimation is applied to SFBC-decoded symbols in order to estimate the noisy time-variant channels, and a frequency domain first order low pass filter is applied to each estimated noisy time-variant channel in order to decrease the Signal-to-Noise Ratio (SNR) requirements that adaptive equalization requires.

As simulation result, the proposed adaptive channel estimation under 120km/hr Jakes' model at 33dB SNR for 2×2 MIMO-OFDM systems have the same performance as One-shot channel estimation. Compare with One-shot channel estimation, the proposed algorithm has reduced 88% multiplier in channel estimate, and the memory size of proposed algorithm is 1/14 of One-shot channel estimations'.

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Chapter 1 Introduction

As mobile communication demand grows, wireless communication systems are pushed to be more mobility. Transmission by means of orthogonal frequency division multiplexing (OFDM) is a spectrally efficient signaling method for communication over frequency-selective fading channels, which divides the given bandwidth into multiple orthogonal subcarriers. Recently, OFDM has been adopted as the downlink transmission scheme for 3GPP Long-Term Evolution (LTE).

Multiple-Input Multiple-Output (MIMO) system with antenna arrays is equipped at both the transmitter and receiver with a signal processor. The multiple transmission antennas at the station in combination with multiple receiver antennas at the mobile can be used to achieve higher peak data rates by enabling multiple data stream transmissions between the station and the mobile by use MIMO spatial multiplexing. Therefore, in addition to large bandwidths and high-order modulations, MIMO spatial multiplexing is used in LTE system to achieve the peak data rate targets. The MIMO spatial multiplexing also provides improvement in cell capacity and throughput as mobile with good channel conditions can benefit from multiple streams transmissions. Therefore, the combination of MIMO and OFDM increase the diversity gain and enhance the system capacity on frequency-selective channels.

The time variation arises because either the transmitter or the receiver is moving and hence the location of reflectors in the transmission path, which gives rise to multipath, will change over time. In mobility system, a well-known effect called Doppler shift [1] reveals when relative motion between transmitter and receiver occurs to. Doppler shift suffer from the related variation of velocity. Unfortunately, the channel estimation is difficult issue in the Doppler shift effect. Channel state information is critical for the performance of time-varying MIMO-OFDM system and is obtained mainly by two ways: one-shot channel estimation based on reference signal and adaptive channel estimation. The disadvantage of one-shot channel estimation needs the large memory to store the channel information. Otherwise, the adaptive channel estimation requires lowers memory.

In this thesis, the contributions are that the novel adaptive channel estimation is proposed and the computing complexity is reduced. The major issues of complexity are memory size and multiplier number.

The effect of Doppler shift is introduced in section 1.1. Section 1.2 describes the time-variant channel effect induced by Doppler shift.

This thesis is organized as follows. Chapter 2 describes the introduction of Long-Term Evolution (LTE) platform. In chapter 3, the adaptive channel estimators are presented. Finally, the simulation results and conclusion are shown in chapter 4 and chapter 5.

1.1 Doppler shift

The relative motion between transmitter and receiver, consider a mobile moving at a constant velocity v, along a path segment having length d between points A and B, while it receives signals from a remote source S, as illustrated in Figure 1-1.



Figure 1-1 Illustration of Doppler effect

The approximate path length traveled by the wave is represented as Δl while the vehicle travels from point A to B, it can be expressed as:

$$\Delta l = d\mathbf{c} \circ \boldsymbol{\theta} = \boldsymbol{\Delta} \quad t \circ \boldsymbol{\theta} \tag{1.1}$$

where Δt is the time required for the vehicle to travel from A to B. Since the source is assumed to be very far away, in order to simplify the deriving of Doppler shift, we assumed that θ at point A and θ' at point B is the same. The approximate amount of phase shift $\Delta \phi$ and frequency shift f_d experienced by the receiver can be described as follows:

$$\Delta \phi = \frac{2\pi\Delta l}{\lambda} = \frac{2\pi v \Delta t}{\lambda} \cos \theta \tag{1.2}$$

$$f_d = \frac{1}{2\pi} \cdot \frac{\Delta \phi}{\Delta t} = \frac{v}{\lambda} \cos \theta.$$
(1.3)

 λ stands for the wave length transmitted from the remote transmitter. Note that the frequency shift f_d and the phase shift $\Delta \phi$ are varying during the travel from point A to B in realistic environment.

1.2 Time-Variant Channel Effect

Multipath propagation with combination of Doppler shift induces time-variant channel effects. As the phenomenon shown in Figure 1-2, a vehicle receives three multipath components while it travels from point A to B. Each multipath component is affected by different time-varying frequency shift and phase shift, therefore the time-variant channel effect reveals.



Figure 1-2 The occasion of time-variant channels

The time-variant channel effect can also be described as a Finite Impulse Response (FIR) filter, which has three time-variant tap gains since there are three multipath components, as shown in Figure 1-3.



Figure 1-3 A 3-tap Finite Impulse Response (FIR) filter model for time-variant

channels

s(t) is the transmitted signal and the received signal $\,y(t)\,$ can be written as a convolution sum

$$y(t) = \sum_{k=0}^{2} w_{k}(t)^{*} \cdot s(t-k)$$
 (1.4)

Each time-variant tap gain $w_k(t)$ is a complex number, which describes the phase and amplitude response of the k^{th} path at time t.

The time-variant tap gains, or Channel Impulse Response (CIR), can be transformed into Channel Frequency Response (CFR), which is also time-variant. Figure 1-4 shows the 15-tap noise-free time-variant CFR at 120km/hr by Jakes' model [2].



Figure 1-4 (a) Amplitude and (b) Phase variations of time-variant CFR by Jakes' model at 120km/hr

Chapter 2 System Platform

3rd Generation Partnership Project (3GPP)[3][4][5] has recently specified an Orthogonal Frequency Division Multiplexing (OFDM) based technology, Evolved Universal Terrestrial Radio Access (E-UTRA), for support of wireless broadband data service up to 300 Mbps in the downlink and 75Mbps in the uplink [6]. In Long Term Evolution (LTE), MIMO technologies have been widely used to improve downlink peak rate, cell coverage, as well as average cell throughput. LTE is not only able to operate in different frequency bands but can also flexibly support different bandwidths.

A unique LTE possibility is to use different UL and DL bandwidths, allowing for asymmetric spectrum utilization. This is possible due to the support of both paired Frequency Division Duplexing (FDD) and unpaired Time Division Duplexing (TDD) band operations. In this thesis we concentrate on TDD.

2.1 LTE PHY Specification

Orthogonal Frequency Division Multiplexing (OFDM) is a popular method for high data rate wireless transmission. OFDM may be combined with antenna arrays at the transmitter and receiver to increase the diversity gain and to enhance the system capacity on time variant and frequency-selective channels, resulting in a multiple-input multiple-output (MIMO) configuration.

2.1.1 Transmitter

The transmitter block diagram of LTE MIMO specified is shown as Figure 2-1.



Figure 2-1 LTE MIMO transmitter

The baseband signal representing a downlink physical channel is defined in terms of the following steps:

- Scrambling of coded bits in each of the code words to be transmitted on a physical channel to prevent a succession of zeros or ones
- Modulation of scrambled bits from some modulation alphabet, e.g. BPSK, QPSK, 16QAM, 64QAM, to generate complex-values modulation symbols.
- Mapping of complex-valued modulation symbols onto one or several transmission layers. Basically a layer corresponds to a spatial multiplexing channel.
- The pre-coder extracts exactly one modulation symbol from each layer, jointly process these symbols, and maps the result in the frequency domain and antenna domain. The space-frequency block code (SFBC) as shown in section 2.2 is used if there has more than one transmit antennas.
- Mapping of complex-valued modulation symbols for each antenna port to the resource elements of the set of resource blocks assigned by the MAC scheduler for transmission of transport block(s).
- Generation of complex-valued time-domain OFDM signal for each antenna port by Inverse Fast Fourier Transform (IFFT). There are 1024 frequency entries for each IFFT, or 1024 sub-carriers in each OFDM symbol. 600 of them are data carriers, the rest 423 are null carriers. Finally, the time domain

signals appended to the Cyclic Prefix (CP) of 144 or 160 sub-carries are transmitted by RF modules.

2.1.2 Receiver

The receiver block diagram of MIMO-OFDM specified in LTE proposal is shown as Figure 2-2.



Fig 2-2 LTE MIMO receiver

The receive signals are first synchronized to recognize each OFDM symbol. Each OFDM symbols shall be processed and de-mapped to coded bits through the following steps:

- Each OFDM symbols is transformed to complex-valued frequency-domain OFDM signal by Fast Fourier Transform (FFT). If the OFDM symbol belongs to reference signal (described in section 2.1.4), then it is used for channel estimation.
- De-mapping complex-valued frequency-domain OFDM signal of each antenna port to complex-valued symbols.
- Complex-valued symbols are decoded by SFBC decoder, which needs the CFR information estimated by the channel estimation block. The SFBC decoder can be seen as the equalizer of MIMO-OFDM systems and it is introduced in section 2.2.
- After SFBC decoder, the spatial streams shall be Layer de-mapped and demodulation to complex-values modulation symbols.
- Finally, the data stream is decoded by de-scrambler.

2.1.3 Frame Structure

Shown as Figure 2-3, frame structure is applicable to TDD. Each radio frame of length $T_f = 307200 \cdot T_s = 10ms$ consists of two half-frames of length $153600 \cdot T_s = 5ms$ each. Each half-frame consists of five sub-frames of length $307200 \cdot T_s = 1ms$.



Figure 2-3 LTE frame format

Physical Downlink Share Channel (PDSCH), Reference Signal (RS) and Physical Downlink Control Channel (PDCCH) compose a sub-frame, shown as Figure 2-4. PDSCH is used for all user data, as well as for broadcast system information. Data is transmitted on the PDSCH in units known as transport blocks. A PDCCH carries a message known as Downlink Control Information (DCI), which includes resource assignments and other control information for a mobile or group of mobiles. RS(s) are known signals which do not carry any data. It can be used to do channel estimation.



Figure 2-4 LTE frame format

The transmitted signal in each slot is described by a resource grid of $L \times M$ subcarrier and N OFDM symbols, where L is downlink bandwidth configuration, M is resource block size in the frequency domain, expressed as a number of subcarriers, and N is the number of OFDM symbols in a downlink slot. Resource blocks are used to describe the mapping of certain physical channels to resource elements. The resource grid structure is illustrated in Figure 2-5. In the thesis, L is 50 blocks, M represents 12 subcarriers, and N denotes 14 OFDM symbols.



2.1.4 Reference Signal

LTE downlink reference signals have three types, Cell-specific reference signals, MBSFN reference signals, and UE-specific reference signals. In this thesis we concentrate on Cell-specific reference signals.

The mapping of downlink reference signals is shown in Figure 2-6. Resource element used for reference signal transmission on any of the antenna ports in a slot shall not be used for any transmission on any other antenna port in the same slot and set to zero.



2.2 Space-Frequency Block Coding

In LTE transmit diversity, the pre-coding is the same as Space-Frequency Block Coding (SFBC) [7], it has been applied to frequency selective channels by using OFDM with cyclic prefix, which transforms a frequency selective channel into flat fading sub-channels. The SFBC coding is done across the sub-carriers inside one OFDM symbol. In 2×2 SFBC, SFBC encodes two components onto two antennas over two different frequencies, these two frequencies are chosen to correspond to adjacent subcarriers. Since channel realizations on adjacent subcarriers may differ, SFBC is more sensitive to large delay spreads.

2.2.1 2×2 SFBC

The block diagram on a transmitter using OFDM and space-frequency block coding is shown in Figure 2-7. The mapping scheme of the data symbols X_n for SFBCs with 2 transmit antennas is shown in Table 2-1. The mapping scheme for SFBCs is chosen such that on the first antenna the original data is transmitted without any modification so that the scheme is compatible to systems without SFBC where the second antenna is not implemented or switched off.

	Antenna 1	Antenna 2
Subcarrier n	X_n	$-X_{n+1}^{*}$
Subcarrier n+1	X_{n+1}	X_n^*
	1896	
$\operatorname{Tx} 1 \begin{array}{ c c } X_1 - X_2^* \end{array}$	$X_3 - X_4^*$	$X_{n-1} - X_n^*$
f_1 f_2 f_3	f_4 f_{n-1} f_n	frequency
Tx 2 $X_2 X_1^*$	$X_4 X_3^*$	$X_n X_{n-1}^*$

Table 2-1 Mapping with space-frequency block code and two transmit antenna

Figure 2-7 Space-frequency block coding in an OFDM transmitter

Two subcarriers x_1 and x_2 are about to be transmitted in two subcarriers periods from two transmit antennas. The SFBC-encoded codeword is shown as

Figure 2- 2-8. At frequency f_1 , x_1 and $-x_2^*$ are transmitted through channels h_1 and h_2 . And at frequency f_2 , x_2 and x_1^* are transmitted through channels h_1 and h_2 , respectively.



Figure 2-8 2×2 SFBC MIMO systems for two receivers

Therefore the received signals \mathbf{r}_1 and \mathbf{r}_2 at time f_1 and f_2 can be expressed as

$$\begin{cases} r_1 = h_1 x_1 - h_2 x_2^* \\ r_2 = h_1 x_2 + h_2 x_1^* \end{cases}$$
(2.1)

Rewrite in matrix form

$$\begin{bmatrix} \mathbf{r}_1 \\ \mathbf{r}_2^* \end{bmatrix} = \begin{bmatrix} \mathbf{h}_1 & -\mathbf{h}_2 \\ \mathbf{h}_2^* & \mathbf{h}_1^* \end{bmatrix}, \quad \text{or } \mathbf{R} = \mathbf{H} \cdot \mathbf{X}$$
(2.2)

The channel matrix H can be estimated using reference signals, hence the transmitted symbols can be calculated by

ALLIN,

$$\begin{bmatrix} \mathbf{x}_1 \\ \mathbf{x}_2^* \end{bmatrix} = \begin{bmatrix} \mathbf{h}_1 & -\mathbf{h}_2 \\ \mathbf{h}_2^* & \mathbf{h}_1^* \end{bmatrix}^{-1} \cdot \begin{bmatrix} \mathbf{r}_1 \\ \mathbf{r}_2^* \end{bmatrix}, \text{ or } \mathbf{X} = \mathbf{H}^{-1} \cdot \mathbf{R}$$
(2.3)

Note that the dash-lined components, the second receiver and the two corresponded channels, are ignored during the derivation of the decoding process, since the decoding method is the same as the first receiver. In simulation platform, two receivers are deployed and the SFBC-decoded symbols for each receiver are averaged before output.

2.3 Channel Models

In this section, the time-variant channel models are introduced, this channel model [2] is the well-known Jakes' mode. It will describe as following.

2.3.1 Jakes' Model

Time-variant channel effect can be modeled as a FIR filter with time-variant tap gains. For Jakes' model, the variance of each tap gain obeys Rayleigh distribution. Figure 2-9 shows an n-tap FIR filter with Rayleigh-distributed tap gains, and the corresponded velocity is 120km/hr, shown for 50ms.



Figure 2-9 FIR filter with Rayleigh-distributed tap gains at 120km/hr

s(t) is the transmitted signal and the received signal $\,y(t)\,$ can be written as a convolution sum

$$y(t) = \sum_{k=0}^{n-1} w_k(t)^* \cdot s(t-k)$$
 (2.4)

The Rayleigh-distributed tap gains $w_k(t)$ can be expressed as the sum of sinusoids [8], that is

$$w_{k}(t) = X_{c}(t)\cos 2\pi f_{c}t + X_{s}(t)\sin 2\pi f_{c}t$$

$$X_{c}(t) = 2\sum_{n=1}^{N_{o}}\cos\beta_{n}\cos 2\pi f_{n}t + \sqrt{2}\cos\alpha\cos 2\pi f_{m}t$$

$$X_{s}(t) = 2\sum_{n=1}^{N_{o}}\sin\beta_{n}\cos 2\pi f_{n}t + \sqrt{2}\sin\alpha\cos 2\pi f_{m}t$$
(2.5)

 f_c is the carrier frequency, and it equals to 2.4GHz in 802.11n. f_m is the maximum Doppler frequency shift described in equation (1.1) for $\theta=0$. N_o is the number of oscillators that generate the sinusoid waveforms of angular frequency β_n . Detail description of the sum of sinusoids is referred to [8].

Chapter 3 The Proposed Algorithm

In this chapter, an adaptive channel estimation technique is proposed. 2×2 SFBC for MIMO-OFDM systems is considered. The proposed algorithm consists of three parts, the channel estimation of initial channel, the decision-feedback adaptive channel estimation for estimating the noisy time-variant channel, and the frequency domain lose-pass filter for decreasing the SNR requirement.

Section 3.1 gives an overview of proposed decision-feedback adaptive channel estimation and the use PDCCH for channel estimation and frequency domain adaptive channel estimation is proposed in section 3.2 and 3.3, the 2×2 SFBC is proposed in 3.4, and the lose-pass filter is discussed in section 3.5.

3.1 Adaptive Channel Estimation Overview

The synchronized time-domain signals are transformed into frequency-domain OFDM symbols by FFT. Figure 3-1 describes the data path after FFT for 2×2 LTE MIMO-OFDM systems.



Figure 3-1 2×2 adaptive channel estimation block diagram

After FFT, OFDM symbols classified as Physical Downlink Control Channel (PDCCH) are used for channel estimation. The 2×2 SFBC decoder operates for every two payload symbols, as shown as r_1 and r_2 above. Then the decoded symbols, x'_1 and x'_2 , are decided by decision unit. The functionality of the decision unit is illustrated in Figure 3-1, it uses QAM constellation for decisioning.



Figure 3-1 Functionality of decision unit (QAM decisioning)

The decided symbols are averaged with which of the second receiver. Δx_1 and Δx_2 are the differences between decided symbols and decoded symbols, and they are used by adaptive channel estimation to obtain the variances of the time-variant channels. After adding up with h_1 and h_2 , the time-variant channels, h_1' and h_2' , are ready to update the channel information stored in channel estimation block. The usage of Low pass filter is to mitigate the influence of Additive White Gaussian Noise (AWGN), since the adaptive channel estimation algorithm is sensitive to noise.

The flow chart for adaptive channel estimation is show below.



Figure 3-3 Adaptive channel estimation flow chart

After use PDCCH to do initial channel estimation, the OFDM symbols can only be one of the following categories: reference signal (RS) for channel estimation, or adaptive channel estimation. If the $N + 1^{th}$ OFDM symbol has RS, use these RS to do interpolation, otherwise, use N^{th} OFDM symbol channel to do adaptive channel estimation and create $N + 1^{th}$ OFDM symbol channel.

3.2 Use PDCCH for Channel Estimation

Before adaptive channel estimation, it is necessary to estimation an OFDM symbol channel as initial channel. In this thesis, the PDCCH is used. Physical Downlink Control Channel (PDCCH) carries scheduling assignments and other control information. A physical control channel is transmitted on an aggregation of one of or several consecutive control channel elements (CCEs), where a control channel element corresponds to 9 resource element groups. The number of resource-element groups not assigned to PCFICH or PHICH is N_{REG} . The CCEs available in the system are numbered from 0 and N_{CCE} –1, where $N_{CCE} = \lfloor N_{REG} / 9 \rfloor$. The PDCCH supports multiple formats as listed in Table 3-1.

PDCCH	Number of	Number of	Number of
format	CCEs	resource-element 1896 groups	PDCCH bits
0	1	9	72
1	2	18	144
2	4	36	288
3	8	72	576

Table 3-1: Supported PDCCH formats

As above describe, PDCCH are known signals which do not carry any data. In transmitted resource block, shows in Figure3-4. The PDCCH are second OFDM symbol and third OFDM symbol. We use the third OFDM symbol and zero forcing to find the channel.



Figure 3-4 PDCCH in resource block

3.3 Frequency Domain Adaptive Channel Estimation

Each noisy time-variant channel estimated by adaptive channel estimation is then filtered by Low pass filter. For 2x2 LTE MIMO-OFDM systems, In an OFDM symbol, have 600 subcarriers. After SFBC decoder, there are 300 subcarriers. We get the 50 subcarriers, show in Figure 3-5, as virtual pilot to be adaptive channel estimation, pass through the SFBC coder and interpolation to 600 subcarriers, and then filtered by Low pass filter.



Figure 3-5 Virtual Pilots

3.4 Adaptive Channel Estimation for 2×2 SFBC

In this section, the adaptive channel estimation algorithm can be seen as a transformation of the variances of SFBC-decoded symbols into the variances of time-variant channels. When we finish the one-shot channel estimation, if without adaptive channel estimation, we should keep a large memory to store the channel information until the step of channel equalizer. In architecture level, it will result huge number of gate count. The adaptive channel estimation algorithm applies to each frequency component individually for MIMO-OFDM systems.

For each receiver, the received signal within one SFBC block contains two subcarriers, r_1 and r_2 , which can be expressed as

$$\begin{cases} r_1 = h_1 x_1 + h_2 (-x_2^*) \\ r_2 = h_1 x_2 + h_2 x_1^* \end{cases}$$
(3.1)

 r_f indicates the received signal at frequency index f within SFBC block and the channel from transmitter j is called h_j , which are assumed to be known. x_1 and x_2 stands for the transmitted 2×2 SFBC codeword which are wanted.

By defining
$$\mathbf{R} \triangleq \begin{bmatrix} \mathbf{r}_1 \\ \mathbf{r}_2^* \end{bmatrix}$$
, $\mathbf{H} \triangleq \begin{bmatrix} \mathbf{h}_1 & -\mathbf{h}_2 \\ \mathbf{h}_2^* & \mathbf{h}_1^* \end{bmatrix}$ and $\mathbf{X} \triangleq \begin{bmatrix} \mathbf{X}_1 \\ \mathbf{X}_2^* \end{bmatrix}$, the received SFBC block

can be expressed in matrix form, that is

$$\begin{bmatrix} \mathbf{r}_1 \\ \mathbf{r}_2^* \end{bmatrix} = \begin{bmatrix} \mathbf{h}_1 & -\mathbf{h}_2 \\ \mathbf{h}_2^* & \mathbf{h}_1^* \end{bmatrix} \cdot \begin{bmatrix} \mathbf{x}_1 \\ \mathbf{x}_2^* \end{bmatrix}, \text{ or } \mathbf{R} = \mathbf{H} \cdot \mathbf{X}$$
(3.2)

Since the channels and the received signals are known, x_1 and x_2 can be decoded by using the corresponded decoding process. It multiplies the received signal matrix R by the inverted channel matrix H^{-1} on the left side. Shown as

$$\begin{bmatrix} x_1 \\ x_2^* \end{bmatrix} = \begin{bmatrix} h_1 & -h_2 \\ h_2^* & h_1^* \end{bmatrix}^{-1} \cdot \begin{bmatrix} r_1 \\ r_2^* \end{bmatrix}, \text{ or } X = H^{-1} \cdot R$$
(3.3)

Due to time-variant channel effect, the channels of the consecutive SFBC block are not consistent with the previous ones. For the consecutive SFBC block, the channels are assumed to be h'_1 and h'_2 , which are defined as

$$\begin{cases} \mathbf{h}_{1}^{'} = \mathbf{h}_{1} + \Delta \mathbf{h}_{1} \\ \mathbf{h}_{2}^{'} = \mathbf{h}_{2} + \Delta \mathbf{h}_{2} \end{cases}$$
(3.4)

 Δh_1 and Δh_2 indicate the channel variance during one SFBC block duration. Therefore, the received consecutive SFBC block in matrix form becomes

$$\begin{bmatrix} \mathbf{r}_{1} \\ \mathbf{r}_{2}^{*} \end{bmatrix} = \begin{bmatrix} \mathbf{h}_{1}^{'} & -\mathbf{h}_{2}^{'} \\ \mathbf{h}_{2}^{'*} & \mathbf{h}_{1}^{'*} \end{bmatrix} \cdot \begin{bmatrix} \mathbf{x}_{1} \\ \mathbf{x}_{2}^{*} \end{bmatrix} = \begin{bmatrix} \mathbf{h}_{1} + \Delta \mathbf{h}_{1} & -\mathbf{h}_{2} - \Delta \mathbf{h}_{2} \\ \mathbf{h}_{2}^{*} + \Delta \mathbf{h}_{2}^{*} & \mathbf{h}_{1}^{*} + \Delta \mathbf{h}_{1}^{*} \end{bmatrix} \cdot \begin{bmatrix} \mathbf{x}_{1} \\ \mathbf{x}_{2}^{*} \end{bmatrix}$$
$$= \begin{bmatrix} \mathbf{h}_{1} & -\mathbf{h}_{2} \\ \mathbf{h}_{2}^{*} & \mathbf{h}_{1}^{*} \end{bmatrix} \cdot \begin{bmatrix} \mathbf{x}_{1} \\ \mathbf{x}_{2}^{*} \end{bmatrix} + \begin{bmatrix} \Delta \mathbf{h}_{1} & -\Delta \mathbf{h}_{2} \\ \Delta \mathbf{h}_{2}^{*} & \Delta \mathbf{h}_{1}^{*} \end{bmatrix} \cdot \begin{bmatrix} \mathbf{x}_{1} \\ \mathbf{x}_{2}^{*} \end{bmatrix}, \qquad (3.5)$$
or R=H · X + Δ H · X
where Δ H $\triangleq \begin{bmatrix} \Delta \mathbf{h}_{1} & -\Delta \mathbf{h}_{2} \\ \Delta \mathbf{h}_{2}^{*} & \Delta \mathbf{h}_{1}^{*} \end{bmatrix}$

Applying the decoding process again, that is, multiplying the inverted channel matrix H^{-1} on both left side of equation (3.5), hence

$$H^{-1} \cdot R = X + H^{-1} \cdot \Delta H \cdot X$$

$$\triangleq X'$$
(3.6)
where $X' \triangleq X_{DF} + \Delta X = \begin{bmatrix} (x_1)_{DF} \\ (x_2)_{DF} \end{bmatrix} + \begin{bmatrix} \Delta X_1 \\ \Delta X_2 \end{bmatrix}$

As shown above, in consequence of the time-variant channel effect, the decoded symbol X' contains a residual part with compared to X. The residual part $\triangle X$ indicates the variance of the decoded symbol due to time-variant channel effect. $\triangle X$ can be obtained if the time-variant channel effect is not strong enough to make for a decision error on the decoded symbol X'. That is, if the decision feedback result of X', which is defined as X_{DF} , is equal to the ideal decoded symbol X, then the relationship between $\triangle X$ and $\triangle H$ can be identified. By observing equation(3.6)

if
$$X=X_{DF}$$

 $\Rightarrow X_{DF}+H^{-1} \cdot \Delta H \cdot X_{DF}=X_{DF}+\Delta X$
 $\Rightarrow H^{-1} \cdot \Delta H \cdot X_{DF}=\Delta X$ (3.7)
 $\Rightarrow \Delta H \cdot X_{DF}=H \cdot \Delta X$
where $\Delta X=X'-X_{DF}$

 \triangle H can be solved since it is the only unknown matrix in equation (3.7). Further

more, by defining
$$\triangle R \triangleq \begin{bmatrix} \triangle r_1 \\ \triangle r_2 \end{bmatrix} = H \cdot \triangle X$$
 and $\triangle R_{\text{conj}} \triangleq \begin{bmatrix} \triangle r_1 \\ -\triangle r_2^* \end{bmatrix}$, equation (3.7) can be

rewritten as

$$\Delta \mathbf{R} = \mathbf{H} \cdot \Delta \mathbf{X} = \Delta \mathbf{H} \cdot \mathbf{X}_{DF}$$

$$\Rightarrow \begin{bmatrix} \Delta \mathbf{r}_{1} \\ \Delta \mathbf{r}_{2} \end{bmatrix} = \begin{bmatrix} \Delta \mathbf{h}_{1} & -\Delta \mathbf{h}_{2} \\ \Delta \mathbf{h}_{2}^{*} & \Delta \mathbf{h}_{1}^{*} \end{bmatrix} \cdot \begin{bmatrix} (\mathbf{x}_{1})_{DF} \\ (\mathbf{x}_{2})_{DF} \end{bmatrix}$$

$$\Rightarrow \begin{bmatrix} \Delta \mathbf{r}_{1} \\ -\Delta \mathbf{r}_{2}^{*} \end{bmatrix} = \begin{bmatrix} (\mathbf{x}_{1})_{DF} & (-\mathbf{x}_{2})_{DF} \\ (\mathbf{x}_{2}^{*})_{DF} & (\mathbf{x}_{1}^{*})_{DF} \end{bmatrix} \cdot \begin{bmatrix} \Delta \mathbf{h}_{1} \\ \Delta \mathbf{h}_{2} \end{bmatrix}$$

$$\Rightarrow \begin{bmatrix} \Delta \mathbf{h}_{1} \\ \Delta \mathbf{h}_{2} \end{bmatrix} = \begin{bmatrix} (\mathbf{x}_{1})_{DF} & (-\mathbf{x}_{2})_{DF} \\ (\mathbf{x}_{2}^{*})_{DF} & (\mathbf{x}_{1}^{*})_{DF} \end{bmatrix} \cdot \begin{bmatrix} \Delta \mathbf{h}_{1} \\ \Delta \mathbf{h}_{2} \end{bmatrix}$$

$$\Rightarrow \begin{bmatrix} \Delta \mathbf{h}_{1} \\ \Delta \mathbf{h}_{2} \end{bmatrix} = \begin{bmatrix} (\mathbf{x}_{1})_{DF} & (-\mathbf{x}_{2})_{DF} \\ (\mathbf{x}_{2}^{*})_{DF} & (\mathbf{x}_{1}^{*})_{DF} \end{bmatrix} \cdot \begin{bmatrix} \Delta \mathbf{r}_{1} \\ -\Delta \mathbf{r}_{2}^{*} \end{bmatrix}$$

$$(3.8)$$

Since $\triangle h_1$ and $\triangle h_2$ are solved, h'_1 and h'_2 can be obtained by equation (3.4). Note that equation (3.8) uses the same process as which used in equation (3.3) for SFBC decoding.

3.5 First Order Low Pass Filter

Low pass filter is used on frequency domain to decrease the SNR requirements that adaptive channel estimation needed since it is sensitive to noise. The functionality of frequency domain Low pass filtering is illustrated below

Assume the samples y(n) must be estimated from measurements x(n), it can be written as

$$y(n) = b(1)x(n) + b(2)x(n-1) + \dots + b(nb+1)x(n-nb)$$

$$-a(2)y(n-1) - \dots - a(na+1)y(n-na)$$
(3.9)

a and b are vectors of low pass filter. It is show as Figure 3-6.



Figure 3-6 Low Pass Filter



Chapter 4 Simulation Results

To evaluate the proposed algorithm, a typical MIMO-OFDM system based on LTE is used as the reference design platform. Performance of the proposed 2×2 adaptive channel estimation under time-variant frequency-selective fast-fading channels is simulated. The parameters used in the simulation platform are: OFDM symbol length is 14 and 600 subcarriers in an OFDM symbol. The major parameters are summarized in Table 4-1.

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Table 4-1 Simulation parameters		
Parameter	896 Value	
Number of taps	6	
Doppler speed	120km/s	
Modulation	64-QAM	
Equalization	Zero-Forcing	
FFT size	1024 Bytes	
Signal Bandwidth	20 MHz	
Subframe size	1 ms	

Figure 4-1 shows the Mean Squared Error (MSE) of the OFDM symbols, if without adaptive channel estimation, the MSE of OFDM symbol channel is poor than that with adaptive channel estimation in non-reference signal OFDM symbols. By the way, MSEs are changed according to the different step-size, as show in figure 4-1. The results show that the best step-size in simulation is 0.1. Figure 4-2 expresses the Bit Error Rate (BER) performance with different step-size of the proposed adaptive channel estimation for 2×2 LTE MIMO-OFDM systems under time-variant channels for 120km/hr Jakes' model.



Figure 4-1 MSE of OFDM symbol with step-size 0, 0.1, 0.15, 0.2



Figure 4-2 BER of OFDM symbol with step-size 0.1, 0.15, 0.2

Figure 4-3 addresses the Bit Error Rate (BER) performance of the proposed adaptive channel estimation 2×2 LTE MIMO-OFDM systems, which use the low pass filter to filter the noisy.



Figure 4-3 BER of OFDM symbol with Low pass filter

Figure 4-4 indicates the Bit Error Rate (BER) performance of the proposed adaptive channel estimation 2×2 LTE MIMO-OFDM systems, which use the different initial channel estimation. The first one is use first OFDM symbol's reference signal, and be interpolation as initial channel, the other one use the PDCCH to estimate the initial channel. The results show that use PDCCH to estimate the initial channel has the better performance.



Figure 4-4 BER of OFDM symbol with two channel estimation methods.

The required SNR for BER of the proposed algorithm among various configurations are summarized in Table 4-3.

Sim	ulation configurations	Jakes'model 120km/hr with step-size 0.1	Jakes' model 120km/hr with step-size 0.15	Jakes' model 120km/hr with step-size 0.2
systems	<i>One-shot</i> channel estimation <i>One-tap</i> compensation	27 dB	27 dB	27 dB
MO-OFDM	<i>Adaptive</i> channel estimation <i>Direct</i> compensation	38 dB	40 dB	45 dB
2×2 MI	Adaptive channel estimation low pass filter compensation	29 dB	30 dB	33 dB

Table 4-3 Required SNR for BER (64-QAM modulation)

As introduction described, the contribution of proposed adaptive channel estimation is computing complexity reduced which compared with one-shot channel estimation. Table 4-2 exhibits the comparison between the one-shot channel and proposed adaptive channel estimation. The result shows that the one-shot channel estimation needs 600×14 storages to store the channel information, but proposed adaptive channel estimation only needs 600 storages. In channel estimation, the multiplier number of one-shot channel estimation needs 650000, but the proposed algorithm which use PDCCH only needs 900, and in adaptive channel estimation, one-shot channel estimation needs zero multiplier and proposed algorithm needs 75000 multipliers. For estimate channel, the total multiplier number of one-shot channel estimation is 650000, and proposed algorithm is 75900. The computing complexity of proposed algorithm is reduced.

Miller

Channel Estimation	One-shot 1	Adaptive (use first OFDM symbol)	Adaptive (use PDCCH)
Storages	600 × 14	600	600
Multiplier (channel Estimation)	650000	1800	900
Multiplier (Adaptive channel Estimation)	0	75000	75000
Total Multiplier	650000	76800	75900

Table 4-2 complexity comparison between two channel estimation

Chapter 5 Conclusion and Future Work

5.1 Conclusion

In this thesis, adaptive channel estimation is proposed to oppose the Doppler effect under high velocity environments, and make full use of the time- and frequency-domain correlation of the frequency response of time-varying multipath fading channels without requirement of accurate channel statistics.

The presented 2×2 adaptive channel estimation scheme achieves. It is a prototype algorithm for SFBC-coded MIMO-OFDM systems. The distance between QAM constellation points dominates the maximum tolerance to time-variant channels; therefore lower modulation scheme provides higher velocity tolerance. Further more, the proposed adaptive channel estimation algorithm shares the same process with SFBC decoding. It is a significant feature in hardware implementation phase.

5.2 Future Work

High QAM constellation like 256-QAM for higher data rate is going to be deployed. As a result, the distance between constellation points shrinks; therefore a robust decision scheme is demanded. Since QAM decisioning simply outputs the closest constellation points to the input as its decision, FEC-feedback decisioning can be considered.

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