Pilot-Aided Angle-Domain Channel Estimation Techniques for MIMO-OFDM Systems

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Abstract—Early multiple-input multiple-output with orthogonal frequency division multiplexing (MIMO-OFDM) channel estimation techniques treat channels as spatially uncorrelated. However, in many situations, MIMO-OFDM channels tend to be spatially correlated, for example, due to limited scattering. For such channels, estimation performance can be improved through exploitation of prior knowledge of the channel spatial correlation, for example, by means of the linear multiple mean square error (MMSE) technique. This knowledge is, however, not always available. As an alternative, we investigate techniques in the angle domain, where the MIMO-OFDM channel model lends itself to a physical interpretation. Our theoretical analysis and simulation results indicate that the proposed angle-domain approximated MMSE (AMMSE) channel estimation technique performs well in terms of the mean square error (mse) for various channel models representing different indoor environments. When a suitable threshold is chosen, we can use the angle-domain most-significant-taps selection technique instead of the angle-domain AMMSE technique to simplify the channel estimation procedure with little performance loss.

Index Terms—Channel estimation, multiple-input multiple-output (MIMO), orthogonal frequency division multiplexing (OFDM), spatial correlation.

I. INTRODUCTION

Recent research trends have shown that combining the multiple-input multiple-output (MIMO) approach with orthogonal frequency division multiplexing (OFDM) can help to achieve spatial diversity and/or space-division multiplexing gain [1]–[4]. As coherent demodulation, which requires and utilizes the knowledge of channel coefficients, can achieve a 3-dB performance gain compared with differential demodulation [5], it is quite commonly adopted in MIMO-OFDM systems. Therefore, accurate and robust channel estimation that permits the realization of coherent demodulation is very important to ensure reliable data recovery.

Generally speaking, pilot-aided channel estimation is based on either the least squares (LS) [6] or the linear MMSE (LMMSE) technique [7], [8]. The essential difference between these two types of techniques is that the channel coefficients are treated as deterministic but unknown constants in the former, and as random variables of a stochastic process in the latter. Compared with LS-based techniques, LMMSE-based techniques yield better performance because they additionally exploit and require prior knowledge of the channel correlation. However, the channel correlation is sometimes not a priori known, which makes LMMSE-based techniques infeasible. To consider wider applications, we will focus on techniques that do not require prior knowledge of the channel correlation.

A typical MIMO-OFDM channel is conceived of as the unique link between the transmitted and noiseless received signals, and is referred to as the array-domain channel. This array-domain channel is treated as spatially uncorrelated in most previous pilot-aided channel estimation techniques for MIMO-OFDM systems (e.g., [4], [9], and [10]), possibly due to the fact that early MIMO studies assume the array-domain channel to be spatially uncorrelated (e.g., [11] and [12]). We call these techniques LS-based techniques in the array domain. However, in many realistic scenarios, the MIMO-OFDM channel tends to be spatially correlated, for example, due to antenna spacing constraints and limited scattering [13]–[15]. For these spatially correlated MIMO-OFDM systems, the LMMSE-based techniques in the array domain, which exploit and require prior knowledge of the channel spatial correlation, yield better performance than LS-based techniques in the array domain [16]–[18]. However, when the channel spatial correlation is not a priori known, which is the assumption made in this paper, these techniques are not applicable. In such cases, to improve the performance of conventional LS-based techniques in the array domain, we investigate techniques in the angle domain, where the channel model lends itself to a simple physical interpretation.

In the angle domain, beamforming patterns with different main lobes are used to characterize the physical propagation environment [19], [20]. For a MIMO system with \( N_t \) transmit and \( N_r \) receive antennas, the beamforming patterns have \( N_t \) transmit lobes and \( N_r \) receive lobes. A pair of transmit and receive lobes forms one angle-domain bin, and, thus, the angle domain is partitioned into \((N_t \times N_r)\) angle-domain bins. For example, as shown in Fig. 1, the transmit lobe 0 together with receive lobe 0 corresponds to the angle-domain bin \((0, 0)\).
Then, multiple unresolvable physical paths (e.g., paths 1 and 2) that occur in the angle-domain bin (0, 0) can be approximately aggregated into one resolvable path, and the paths from other directions (e.g., paths 3 and 4) will have little effect on this resolvable path because they originate or end at other lobes. Consequently, different physical paths approximately contribute to different angle-domain bins, and the channel coefficients in different angle-domain bins can be assumed to be approximately spatially uncorrelated. Furthermore, when some angle-domain bins contain few physical paths due to limited scattering, the corresponding channel coefficients should approach zero. Based on these two special properties for the angle-domain channel coefficients, we introduce three novel channel estimation techniques for MIMO-OFDM systems in this paper.

Note that in MIMO-OFDM systems, we classify the signals and channels in two domain types. The first type is represented by either array or angle domain, and the latter one is represented by either time or frequency domain. The channel correlation in the two domain types is referred to as the channel spatial correlation and the channel frequency correlation, respectively. Additionally, we use the term channel correlation for both the channel spatial and frequency correlation in this paper. Hereinafter, we will explicitly state which representation is used for each domain type. For example, the angle–time domain means that the angle and time representations are used for the above two domain types, respectively. When we state the representation for only one domain type, we mean that both representations for the other domain type are applicable unless indicated otherwise. For example, the angle domain refers to either the angle–time or angle–frequency domain.

We focus on angle-domain channel estimation techniques in this paper. More specifically, we consider the angle–frequency and angle–time domain techniques. In the angle–frequency domain, when some angle–frequency domain bins contain few physical paths due to limited scattering, the corresponding channel coefficients should approach zero. This allows us to choose a suitable threshold for ignoring the small-valued channel taps and retaining only the most significant taps (MST) to reduce the effect of noise on the estimates, thereby improving the performance of the channel estimation technique. We call this technique of retaining channel coefficients of sufficiently large power as the MST selection technique. In the angle–time domain, we can also use the MST selection technique. Furthermore, the channel coefficients in the angle–time domain are approximately uncorrelated. Thus, we may use the channel power instead of the channel correlation to approximately perform the angle–time domain LMMSE technique when the SNR is known or reliably estimated. We call the resulting technique the approximated MMSE (AMMSE) technique. Note that we refer to the channel power as the channel average power in this paper. When the channel power is not available, we can use the channel instantaneous power (i.e., the instantaneous power of estimated channel coefficients) to estimate the channel power. In such cases, to maintain the estimated channel power positive and the estimation reliable, a threshold is required to ignore coefficients with low instantaneous power. Furthermore, we will not consider the AMMSE technique in the angle–frequency domain because the channel correlation in the corresponding frequency domain might be too high to be reasonably replaced by the channel power.

To our best knowledge, this paper is the first to systematically study angle-domain channel estimation techniques for MIMO-OFDM systems. We note that one angle-domain channel estimation technique has been investigated for single-input multiple-output OFDM systems [23]. In this technique, the angle–time domain bins that correspond to the identical angular lobes (but with different time indexes) are grouped into one angle–time domain beam. Examining all the estimated channel coefficients in one beam, only the beams that contain significant peak values along the time axis are identified as the signal beams and, thus, retained. All the remaining beams are considered as noise (or interference) beams and are ignored. This technique was shown to be effective when the channel power is concentrated in a few beams. However, when the channel power is distributed over all the beams (e.g., the channel model E in [24]), this technique may hardly improve the performance. To overcome this problem and investigate the techniques for MIMO-OFDM systems, we do not group the angle–time domain bins into beams. Instead, we independently filter the noise in each angle–time domain bin based on the fact that the multipath components are approximately disjoint in the angle–time domain. Then, as introduced, we use either the MST selection technique or the AMMSE technique to estimate channels in the angle–time domain. Furthermore, we also investigate the MST selection technique in the angle–frequency domain.
The three angle-domain channel estimation techniques proposed in this paper have two main advantages. First, the achieved performance gain over the conventional LS technique does not require prior knowledge of the channel correlation and even the channel power. Therefore, the techniques are applicable to various propagation environments. Second, they can use conventional array-domain estimators as the coarse estimators and perform postprocessing in the angle domain. Thus, they are well suited for chip design because integrating these estimators will not significantly change the existing system architecture. It should be noted that materialization as a postprocessor is not a requirement for the angle-domain techniques. They can be also directly implemented in the angle domain by first transforming the transmitted and received signals from the array domain into the angle domain. However, this may modify the existing system architecture and, thus, will not be a focus in this paper. Instead, we concentrate on describing the angle-domain techniques in the framework of a postprocessor.

The major contributions and results of this paper are as follows.

1) We systematically develop the channel estimation techniques in the angle–time and angle–frequency domains for MIMO-OFDM systems. We find that the proposed techniques perform particularly well in the angle–time domain.

2) We develop a unified approach to analyze the performance of the MST and AMMSE channel estimation techniques in terms of the mse. Based on this approach, we also develop a simple way to compare the performance of different angle-domain techniques with the help of the first derivative test [25].

3) The performance of all the angle-domain techniques is dependent on the respective thresholds. Nevertheless, we find that setting the threshold to be two times of noise variance is sufficient for the angle–time domain MST selection and AMMSE techniques to yield better performance than the conventional LS technique at all the SNRs for various IEEE 802.11n channel models [24].

4) Of all the proposed angle-domain techniques, our theoretical analysis and simulation results demonstrate that the angle–time domain AMMSE technique results in the best performance and achieves up to 8-dB performance gain when the mse is $10^{-2}$ compared to the conventional LS technique.

Throughout this paper, we make four assumptions. First, all the transmit and receive antennas have the same polarization and radiation patterns, and the geometry of the antenna array is the uniform linear array. Second, the spacing between the antennas at the transmitter or the receiver is much smaller than the distances between the scatterers and antenna arrays so that the paths from a given scatterer to all the transmit or receive antennas are approximated to be parallel. This is a typical assumption for the analysis of MIMO-OFDM systems [20]–[22]. Third, as our main concern is the indoor propagation environment, we assume the channel to be time invariant over a given training period. Fourth, the channel correlation and even the channel power are not available to the receiver.

This paper is organized as follows. Section II describes conventional MIMO-OFDM systems. In Section III, the angle-domain representation of MIMO-OFDM systems is presented. In Section IV, three angle-domain techniques are proposed to estimate the MIMO-OFDM channels, and their performance is analyzed in Section V. We evaluate the performance of different techniques in Section VI by simulating typical IEEE 802.11n channel models. Last, we conclude this paper in Section VII.

II. MIMO-OFDM SYSTEM MODEL

In this section, we first introduce the array-domain representation of MIMO-OFDM systems. Therefore, for the sake of convenience, we refer to the time and frequency domains as the array–time and array–frequency domains, respectively, in this section. Then, in the next section, we describe the angle-domain representation of MIMO-OFDM systems.

In a typical MIMO-OFDM system with $N_t$ transmit and $N_r$ receive antennas, the high rate symbols to be transmitted are first grouped into blocks of $N_d$ data symbols at the transmitter. These groups are called the frequency-domain OFDM symbols, and the $n$th group at the $(i_r,i_t)$th transmitter is represented by the vector $x_{i_r,i_t}(n) = [x_{i_r,i_t}(0,n), x_{i_r,i_t}(1,n), \ldots, x_{i_r,i_t}(N_d-1,n)]^T$, where $i_r$ and $n$ denote the indexes of the transmitter and the OFDM symbol, respectively. Next, an inverse discrete Fourier transform (IDFT) block is applied to each OFDM symbol at each transmitter. The IDFT block at the transmitter and the discrete Fourier transform (DFT) block at the receiver serve to modulate and demodulate the data on the orthogonal subcarriers, respectively. At the IDFT output (i.e., in the time domain), a cyclic prefix (CP) of length $N_g$, as a copy of the last part of the current OFDM symbol, is inserted at the beginning of each symbol to avoid ISI, and its length $N_g$ is assumed to be not shorter than the channel length. The resulting $n$th time-domain OFDM symbol at the $(i_r,i_t)$th transmitter is represented by

$$s_{i_r,i_t}(n) = \sum_{l=0}^{N_d-1} x_{i_r,i_t}(l,n)e^{j2\pi(l-N_g)/N_d}$$

for $m = 0, 1, \ldots, N_d + N_g - 1$.

The samples $\{s_{i_r,i_t}(m,n)\}$ are sent through a frequency-selective fading channel, which can be represented by an equivalent discrete-time linear finite duration channel impulse response (CIR) given by a sequence of channel matrices $C(l)$ for $l = 0, 1, \ldots, N_h - 1$, where $N_h$ is the time span of the MIMO channel, and $C(l)$ is an $N_r \times N_t$ matrix whose $(i_r,i_t)$th element $c_{i_r,i_t}(l)$ represents the channel coefficient from the $(i_r,i_t)$th transmit antenna to the $(i_r,i_t)$th receive antenna at delay $l$. Assuming that the transmitter and the receiver are perfectly synchronized, the received OFDM symbols become free of ISI when the CP is removed from each symbol at all the

\footnote{For the sake of simple description, $C(l)$ represents the channel matrix at each integer time index. This representation is obvious for the sample-spaced channels. For the non-sample-spaced channels, $C(l)$ can be obtained via the interpolation, as shown in [6].}
By performing an $N_d$-point DFT on the resulting symbols at each receiver, we obtain the corresponding frequency-domain symbols. The resulting frequency-domain data sample at the $k$th subcarrier and $(i_r)$th receiver in the $n$th OFDM symbol is given by

$$y_{i_r}(k, n) = \sum_{i_t=0}^{N_t-1} h_{i_r,i_t}(k)x_{i_t}(k, n) + \vartheta_{i_r}(k, n) \quad (2)$$

where $h_{i_r,i_t}(k)$ is the channel transfer function (CTF) at the $k$th subcarrier from the $(i_t)$th transmitter to the $(i_r)$th receiver, and the noise $\vartheta_{i_r}(k, n)$ is assumed to be additive white Gaussian with variance $\sigma^2_r$. In a compact notation, the received block of the $n$th OFDM symbols at the $k$th subcarrier can be written as

$$y(k, n) = H(k)x(k, n) + \vartheta(k, n) \quad (3)$$

where $y(k, n) = [y_0(k, n), y_1(k, n), \ldots, y_{N_r-1}(k, n)]^T$, $x(k, n) = [x_0(k, n), x_1(k, n), \ldots, x_{N_r-1}(k, n)]^T$, and

$$H(k) = \begin{bmatrix}
h_{0,0}(k) & h_{0,1}(k) & \cdots & h_{0,N_r-1}(k) \\
h_{1,0}(k) & h_{1,1}(k) & \cdots & h_{1,N_r-1}(k) \\
\vdots & \vdots & \ddots & \vdots \\
h_{N_r-1,0}(k) & h_{N_r-1,1}(k) & \cdots & h_{N_r-1,N_r-1}(k) 
\end{bmatrix}$$

is the CTF matrix at the $k$th subcarrier. Furthermore, the received $n$th OFDM signal in the time domain is given by

$$z(m, n) = \sum_{l=0}^{N_s-1} C(l)s(m - l, n) + u(m, n) \quad (4)$$

where $s(m, n) = [s_0(m, n), s_1(m, n), \ldots, s_{N_r-1}(m, n)]^T$, $z(m, n) = [z_0(m, n), z_1(m, n), \ldots, z_{N_r-1}(m, n)]^T$, and $u(m, n)$ are the transmitted, received, and noise vectors, respectively, at the $n$th sample of the $n$th OFDM symbol.

In this section, we represent MIMO-OFDM systems in the angle domain. This is an extension of the work for MIMO flat-fading systems, as shown in [20].

As the channel models for MIMO-OFDM systems are commonly introduced in the array–time domain [24], [26], we start to represent the corresponding angle–time domain MIMO-OFDM systems from (6). Suppose that there is an arbitrary number of physical paths (four paths are illustrated in Fig. 1) between the transmit and receive antennas at time $t$; the $i$th path has an attenuation of $a_i$ with an angle $\phi_i$ ($\Omega_i^t := \sin \phi_i$) and $\phi_i^r$ ($\Omega_i^r := \sin \phi_i^r$) for the transmit and receive antennas, respectively. Then, $C(l)$ is given by

$$C(l) = \sum_i a_i^b e_r(\Omega_i^t) e_t^H(\Omega_i^r) \quad (8)$$

where

$$a_i^b := a_i \sqrt{N_tN_r} \exp \left( \frac{j2\pi d_i}{\lambda_c} \right) \quad (9)$$

$$e_r(\Omega_i^t) := \frac{1}{\sqrt{N_t}} \begin{bmatrix} 
1 \\
\exp [j(2\pi \Delta_t \Omega_i^t)] \\
\vdots \\
\exp [j(N_r - 1)(2\pi \Delta_t \Omega_i^t)]
\end{bmatrix} \quad (10)$$

$$e_t(\Omega_i^r) := \frac{1}{\sqrt{N_r}} \begin{bmatrix} 
1 \\
\exp [j(2\pi \Delta_r \Omega_i^r)] \\
\vdots \\
\exp [j(N_r - 1)(2\pi \Delta_r \Omega_i^r)]
\end{bmatrix} \quad (11)$$

where the superscript $H$ denotes the Hermitian transpose; $\{N_t, \Delta_t, e_r(\Omega_i^t)\}$ and $\{N_r, \Delta_r, e_t(\Omega_i^r)\}$ are the number of antennas, the separation between adjacent antennas normalized by $\lambda_c$, and the array response vectors, respectively, for the transmit and receive antennas, respectively; $\lambda_c$ is the carrier wavelength; and $d_i$ is the distance between the last transmit and receive antennas along path $i$.

As from [20], the orthonormal bases for the angle–time transmitted and received signals are given by

$$\xi_t := \left\{ e_t(0), e_t \left( \frac{1}{L_t} \right), \ldots, e_t \left( \frac{N_t - 1}{L_t} \right) \right\} \quad (12)$$

$$\xi_r := \left\{ e_r(0), e_r \left( \frac{1}{L_r} \right), \ldots, e_r \left( \frac{N_r - 1}{L_r} \right) \right\} \quad (13)$$

respectively, where $L_t = N_t \Delta_t$ and $L_r = N_r \Delta_r$ are the normalized antenna array lengths of the transmitter and the receiver, respectively. Let $U_t$ and $U_r$ be the unitary matrices whose columns are the basis vectors in (12) and (13), respectively. Then, we can transform the $n$th samples of the $n$th

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3For notational convenience, we ignore the time index $l$ in these three variables. We also assume that the fractional bandwidth is small such that $\lambda_c, e_r(\Omega_i^t)$, and $e_t(\Omega_i^r)$ are approximated to be unchanged over the whole signal bandwidth.
transmitted and received OFDM symbol from the array–time domain into the angle–time domain by
\[ s^a(m, n) := U^H_t \bar{s}(m, n) \]
\[ z^a(m, n) := U^H_t \bar{z}(m, n) \]
respectively, where the superscript “a” denotes the angle-domain variables. From (4), we obtain the angle–time domain MIMO-OFDM system equation as
\[ z^a(m, n) = \sum_{l=0}^{N_a-1} C^a(l) s^a(m-l, n) + u^a(m, n) \]
(16)
where
\[ C^a(l) := \begin{bmatrix} c_{0,0}^a(l) & c_{0,1}^a(l) & \cdots & c_{0,N_a-1}^a(l) \\ c_{1,0}^a(l) & c_{1,1}^a(l) & \cdots & c_{1,N_a-1}^a(l) \\ \vdots & \vdots & \ddots & \vdots \\ c_{N_a-1,0}^a(l) & c_{N_a-1,1}^a(l) & \cdots & c_{N_a-1,N_a-1}^a(l) \end{bmatrix} \]
(17)
is the angle–time domain channel matrix, and
\[ u^a(m, n) = U^H_t u(m, n) \]
(18)
is the angle–time domain noise vector that still satisfies an independent identically distributed multivariate complex normal distribution. As explained in [20], due to the finite number of antennas, multiple unresolved physical paths can be appropriately aggregated into one resolvable path with gain \( c_{p,k_l}^a(l) \). This gain uniquely corresponds to the channel coefficient for the \((k_r, k_i)\) angle-domain bin at delay \( l \). Hence, different physical paths (approximately) contribute to different elements of \( C^a(l) \). This means that the angle–time domain channel matrix \( C^a(l) \) lends itself to a physical interpretation. For example, in Fig. 1, paths 1 and 2 contribute to the element \( c_{0,0}^a(l) \). Paths 3 and 4 contribute to the elements \( c_{1,3}^a(l) \) and \( c_{2,2}^a(l) \), respectively. All the other elements of \( C^a(l) \) approach zero because no paths contribute to these elements.

By analogy to the angle–time domain case, from (3), we obtain the angle–frequency domain MIMO-OFDM system equation at the \( k \)th subcarrier as
\[ y^a(k, n) = H^a(k) x^a(k, n) + \vartheta^a(k, n) \]
(19)
where
\[ x^a(k, n) := U^H_t x(k, n) \]
(20)
\[ y^a(k, n) := U^H_t y(k, n) \]
(21)
\[ \vartheta^a(k, n) := U^H_t \vartheta(k, n) \]
(22)
\[ H^a(k) := U^H_t H(k) U_t \]
(23)

IV. ANGLE-DOMAIN CHANNEL ESTIMATION

As the physical path within one angle–time domain beam has most of its energy within this beam, the elements within one angle-domain channel matrix \( C^a(l) \) or \( H^a(k) \) maintain low spatial correlation. Furthermore, various elements in the angle-domain channel matrices tend to approach zero due to limited scattering. Therefore, we may independently perform noise filtering in each angle–time or angle–frequency domain bin to improve the estimation performance. The angle-domain channel can be directly estimated in the angle domain [27] by first transforming the transmitted and received signal from the array–frequency domain into the angle–frequency domain as shown in (20) and (21). However, this may modify the existing system architecture and, thus, will not be a focus in this paper. Instead, without affecting the estimation performance, we concentrate on describing the angle-domain techniques in the framework of a postprocessor. This follows three steps: 1) performing the coarse channel estimation in the array domain; 2) in the postprocessor, transforming the estimated channel from the array domain into the angle–time or angle–frequency domain where the noise filtering process in each angle-domain bin is independently performed; and 3) transforming back the filtered estimated channel into the array domain.

In the literature, there exist two types of frequency-domain pilot arrangements in OFDM systems for the implementation of pilot-aided channel estimation techniques—the block-type and comb-type pilot arrangements. The first type is realized by inserting pilots into all of the subcarriers within a periodical time interval, and the estimated channel coefficients that are obtained from one period are used for further signal processing (e.g., equalization, detection, etc.) until the next period of pilots is received. This type of pilot arrangement is particularly suitable for slowly fading channels. The second type is performed by inserting pilots into a certain number of subcarriers of each OFDM symbol and using the estimated channel coefficients at these pilot subcarriers to interpolate the channel coefficients at other subcarriers. Compared with the first type, this type is more suitable for rapidly time-varying channels in that it can track time variation within each OFDM symbol. However, it may suffer an irreducible estimation error floor due to the interpolation.

In typical MIMO-OFDM systems, only one transmit antenna sends pilot symbols in a given time or frequency position [3], [10], [28]. This technique can lead to very simple MIMO-OFDM channel estimation because the received signals at a given time or frequency position correspond to one unique channel coefficient from a given transmit antenna to a given receive antenna. In such cases, the channel estimation for a MIMO-OFDM system with \( N_t \) transmit and \( N_r \) receive antennas becomes the channel estimation for the total \( N_t \times N_r \) single-input single-output (SISO) OFDM systems. Thus, the well-developed SISO-OFDM channel estimation techniques (e.g., [6], [7], [29], and [31]) are directly applicable to the estimation of MIMO-OFDM channels. In this paper, we use the conventional array–frequency domain LS technique [6] to coarsely estimate the array–frequency domain \( h_{r,t}\) (\( k \)) because the knowledge of channel correlation is assumed to be not available. For the sake of a simple description, as shown in Fig. 2, we assume that pilots from different transmit antennas are time orthogonal to each other. Thus, only one transmit antenna is used to transmit pilots in each OFDM training
symbol period as introduced in [3]. The proposed techniques can be easily extended to other pilot transmission schemes (e.g., [28]) but will not be covered in this paper.

A. Angle–Frequency Domain Technique

From (3) and (7) and using the assumption that the pilots from different transmit antennas are time orthogonal to each other, we obtain

\[ \mathbf{Y} = \mathbf{Xh} + \tilde{\mathbf{\theta}} \]  

(24)

where \( \mathbf{Y} = [\mathbf{Y}^T(n), \mathbf{Y}^T(n+N_1), \ldots, \mathbf{Y}^T(n+(N_c-1)N_1)]^T \) with \( \mathbf{Y}^T(n) = [\mathbf{y}^T(0,n), \mathbf{y}^T(0,n+1), \ldots, \mathbf{y}^T(0,n+N_t-1), \mathbf{y}^T(1,n), \mathbf{y}^T(1,n+N_t-1), \ldots, \mathbf{y}^T(N_d-1,n), \mathbf{y}^T(N_d-1,n+N_t-1)]^T \), \( \mathbf{X} = [\mathbf{X}(n), \mathbf{X}(n+N_1), \ldots, \mathbf{X}(n+(N_c-1)N_1)]^T \) with \( \mathbf{X}(n) = \text{diag} \{ \mathbf{x}^T(0,n), \mathbf{x}^T(0,n+1), \ldots, \mathbf{x}^T(0,n+N_t-1), \mathbf{x}^T(1,n), \ldots, \mathbf{x}^T(1,n+N_t-1), \ldots, \mathbf{x}^T(N_d-1,n), \mathbf{x}^T(N_d-1,n+N_t-1) \} \), and \( \tilde{\mathbf{\theta}} = [\tilde{\mathbf{\theta}}^T(n), \tilde{\mathbf{\theta}}^T(n+N_1), \ldots, \tilde{\mathbf{\theta}}^T(n+(N_c-1)N_1)]^T \) with \( \tilde{\mathbf{\theta}}^T(n) = [\tilde{\mathbf{\theta}}^T(0,n), \tilde{\mathbf{\theta}}^T(0,n+1), \ldots, \tilde{\mathbf{\theta}}^T(0,n+N_t-1), \tilde{\mathbf{\theta}}^T(1,n), \ldots, \tilde{\mathbf{\theta}}^T(1,n+N_t-1), \ldots, \tilde{\mathbf{\theta}}^T(N_d-1,n), \tilde{\mathbf{\theta}}^T(N_d-1,n+N_t-1)]^T \) are the received signal vector, the transmitted signal matrix, and the noise vector, respectively, and \( N_c \) is the number of pilots used for each channel coefficient in the LS channel estimation. Note that \( \mathbf{X} \) is a block diagonal matrix. Then, the array–frequency domain LS estimator is given by

\[ \mathbf{h}_{\text{LS}} = (\mathbf{X}^H \mathbf{X})^{-1} \mathbf{X}^H \mathbf{Y}. \]  

(25)

By rearranging the vector form \( \mathbf{h}_{\text{LS}} \) into its matrix form, in this first step, we obtain the coarsely estimated array–frequency domain channel matrix \( \mathbf{H}(k) \) for \( k = 0, 1, \ldots, N_d-1 \).

In the second step, we transform the estimated array–frequency domain channel matrix from the array–frequency domain \( \mathbf{H}(k) \) into the angle–frequency domain \( \tilde{\mathbf{H}}(k) \) by the use of (23) for all the subcarriers under consideration. Let \( \tilde{h}_{i_r,i_l}(k) \) denote the \((i_r,i_l)\)th element of \( \tilde{\mathbf{H}}(k) \). Then, the filtered angle–frequency domain channel coefficient is given by comparing the power of \( \tilde{h}_{i_r,i_l}(k) \) with a threshold \( \eta \) as follows:

\[ \tilde{h}_{i_r,i_l,\text{MST}}(k) = \begin{cases} 
\tilde{h}_{i_r,i_l}(k), & \text{if } |\tilde{h}_{i_r,i_l}(k)|^2 \geq \eta \\
0, & \text{otherwise}
\end{cases} \]  

(26)

Last, the filtered angle–frequency domain channel matrices are transformed back to the array–frequency domain. We call this technique the angle–frequency domain MST selection technique.

To make a fair complexity comparison among all the techniques, we assume that LS-estimated array–frequency domain channel coefficients are available beforehand. Then, from (23), obtaining the angle–frequency domain \( \tilde{\mathbf{H}}(k) \) from the array–frequency domain \( \mathbf{H}(k) \) requires \( N_t + N_r \) complex multiplication operations for each channel coefficient. Furthermore, transforming the estimated angle–frequency domain into the array–frequency domain also requires \( N_t + N_r \) complex multiplication operations for each channel coefficient. Therefore, the total required number of complex multiplication operations for each channel coefficient is \( 2(N_t + N_r) \).

B. Angle–Time Domain Techniques

In many cases such as the nonline-of-sight (NLOS) scenario, the mean angle of departure (AoD) and angle of arrival (AoA) of the clusters of multipath components tend to be uniformly distributed over all angles [26], [33]. Thus, when the number of clusters is relatively large (e.g., the channel model E in [24]), the above angle–frequency domain channel estimation technique may hardly improve over the conventional array–frequency domain LS technique because nearly all the elements of angle–frequency domain channel matrices may not approach zero. As the clusters of multipath components are disjoint in the angle–time domain, we may perform the noise filtering in this domain instead of the angle–frequency domain.

For the implementation of angle–time domain techniques, we first transform the estimated channel coefficients from the array–frequency domain into the array–time domain by the use of DFT [31]. Then, we transform the channel matrices into the angle–time domain by using (17). In the angle–time domain, we can select the MST in channel matrices to reduce the effect of noise on the estimates. Now, the estimated angle–time domain channel coefficient becomes

\[ \tilde{c}_{i_r,i_l,\text{MST}}(l) = \begin{cases} 
\tilde{c}_{i_r,i_l}(l), & \text{if } |\tilde{c}_{i_r,i_l}(l)|^2 \geq \eta \\
0, & \text{otherwise}
\end{cases} \]  

(27)

where \( \tilde{c}_{i_r,i_l}(l) \) is the coarsely estimated angle–time domain channel coefficient.

Note that we presume that the channel spatial correlation is not available to the receiver in this paper. Therefore, the conventional LMMSE technique that utilizes the channel spatial correlation is not applicable here. However, as discussed, the channel coefficients in the angle domain at a given time are approximately spatially uncorrelated. Therefore, we may use the channel instantaneous power to approximate the channel correlation (here, the channel power is also assumed to be not available). As the approximated channel correlation matrix is a diagonal matrix, the LMMSE technique that jointly filters all the channel coefficients becomes the independent spatial filtering for each channel coefficient. Furthermore, the channel coefficient is uncorrelated to the noise. Therefore, we may estimate the channel instantaneous power for each coefficient as \( |\tilde{c}(l)|^2 - \sigma_f^2 \). The noise variance \( \sigma_f^2 \) is assumed to be known here. In practice, it can be estimated during periods when no
transmitted signal is detected at or virtual carriers where no data are transmitted. Then, the angle–time domain AMMSE technique is realized as

$$
\tilde{e}^{a}_{t_r,t_i,\text{AMMSE}}(l) = \begin{cases} 
\frac{|\tilde{e}^{a}_{t_r,t_i}(l)|^2 - \sigma^2_f}{\sigma^2_f} \tilde{e}^{a}_{t_r,t_i}(l), & \text{if } |\tilde{e}^{a}_{t_r,t_i}(l)|^2 \geq \eta \\
0, & \text{otherwise.}
\end{cases}
$$

(28)

Here, the threshold \( \eta \) is usually chosen to be smaller than \( \sigma^2_f \). Otherwise, the approximated channel power \( (|\tilde{e}^{a}_{t_r,t_i}(l)|^2 - \sigma^2_f) \) becomes negative. In comparison with the MST selection technique, the difference is the use of a dynamic multiplication factor instead of a constant one. The factor turns out to be crucial in improving the performance of channel estimation, as shown in the following two sections. However, the complexity is only increased by only an additional complex multiplication for each channel coefficient.

After the noise filtering in the angle–time domain either by (27) or (28), we transform back the estimated channel into the array–time domain and then into the array–frequency domain.

Similar to the angle–frequency domain MST technique, we assume that LS-estimated array–frequency domain channel coefficients are available beforehand to make a fair complexity comparison. Then, from (17), we know that the transformations between the array–time domain and the angle–time domain require a total of \( 2(N_r + N_t) \) complex multiplication operations for each channel coefficient. In addition, the angle–time domain techniques require the transformations between the array–frequency domain and the angle–time domain. This requires a total of \( 2N_d \) complex multiplication operations for each channel coefficient. Typically, \( N_d \) is a power of 2. Then, using the FFT and the IFFT [35] for transformations between the array–time and array–frequency domains, the total complex multiplication operations required for each channel coefficient is reduced to \( \log_2 N_d N_r \) complex multiplication operations for each channel coefficient. From the above, the total required complex multiplication operations for each channel coefficient in the angle–time domain MST selection technique is \( 2(N_r + N_t) + \log_2 N_d \). For the angle–time domain AMMSE technique, an additional complex multiplication is needed. Therefore, the total required complex multiplication operations for each channel coefficient in the angle–time domain AMMSE technique is \( 2(N_r + N_t) + 1 + \log_2 N_d \). Compared with the angle–frequency domain MST technique, these angle–time domain techniques always have higher complexity. Nevertheless, our results show that these angle–time domain techniques always outperform the angle–frequency domain MST technique for all the channel models under consideration.

Note that the angle–time domain MST selection and AMMSE techniques are well suited for the sample-spaced channels. The performance will be degraded for the nonsample-spaced channels due to the power leakage [6]. Nevertheless, our results show that the angle–time domain techniques still outperform the array–frequency domain LS technique for all the channel models under consideration.

V. PERFORMANCE ANALYSIS OF ANGLE-DOMAIN CHANNEL ESTIMATION TECHNIQUES

For channel estimation techniques, one of the most important performance measures is the mse, which measures the average mean squared deviation of the estimator from the true value [30]. In this section, we present a unified approach to computing the mse of the angle-domain (i.e., either angle–time or angle–frequency domain) MST selection and AMMSE techniques. Note that the array–time and array–frequency domains are related by a unitary transformation. Thus, of a given estimation technique, the mse represented in either the array–time or array–frequency domain yields the same result and is given by

$$
mse = \frac{1}{N_d N_t N_r} E \left[ \| \hat{\mathbf{h}} - \mathbf{h} \|^2 \right]
$$

(29)

where \( \mathbf{h} \) is either \( \mathbf{h} \) or \( \mathbf{c} \), and \( \mathbf{h} \) is the estimated \( \mathbf{h} \).

Here, we use the array–frequency domain LS technique to perform the coarse channel estimation. Then, the resulting mse of the coarse estimation is given by

$$
mse_{LS} = \frac{1}{N_d N_t N_r} E \left[ \| \overline{\mathbf{h}}_{LS} - \mathbf{h} \|^2 \right]
$$

(30)

In many cases such as in the IEEE 802.11a standard [32], the powers of all pilots are unity. Therefore, we also assume that \( E[(X^H X)^{-1}] \) is the identity matrix. Then, since \( X \) is diagonal, (30) becomes

$$
mse_{LS} = \sigma^2_f.
$$

(31)

Let \( \mathbf{h}^a \) represent the stacked angle-domain channel vector. Then, from (23), we obtain

$$
\mathbf{h}^a = \mathbf{B} \overline{\mathbf{h}}
$$

(32)

where

$$
\mathbf{B} = \mathbf{I}_{N_d} \otimes \left( \left( \mathbf{I}_{N_t} \otimes \mathbf{U}_r^H \right) \left( \mathbf{U}_r^T \otimes \mathbf{I}_{N_t} \right) \right)
$$

(33)

is an \( (N_d N_t N_r \times N_d N_t N_r) \) matrix, and the superscript \( T \) denotes the transpose. It is easily verified that the matrix \( \mathbf{B} \) is unitary, i.e.,

$$
\mathbf{B}^H \mathbf{B} = \mathbf{I}_{N_t \times N_t}.
$$

(34)

Then, the angle-domain MST selection or AMMSE technique is given by

$$
\overline{\mathbf{h}}^a = \mathbf{M} \mathbf{h}^a_{LS}
$$

(35)

where \( \mathbf{M} \) is a diagonal \( (N_d N_t N_r \times N_d N_t N_r) \) matrix that represents either the MST selection or AMMSE process, and \( \mathbf{h}^a_{LS} \) is the stacked LS-estimated channel vector in the angle domain. Note that the \( i \)th diagonal element of \( \mathbf{M} \) (denoted as \( m_i \)) is dependent on the channel coefficient and the noise.
Denote \( \hat{r}_i \) as the instantaneous power of the \( i \)th element of \( \mathbf{h}_\text{LS} \).

In the MST selection techniques, we have

\[
m_i = \begin{cases} 
1, & \text{if } \hat{r}_i \geq \eta \\
0, & \text{otherwise}. 
\end{cases}
\] (36)

In the AMMSE technique, from (28), we have

\[
m_i = \begin{cases} 
\frac{\hat{r}_i - \sigma_i^2}{\hat{r}_i}, & \text{if } \hat{r}_i \geq \eta \\
0, & \text{otherwise} 
\end{cases}
\] (37)

where the threshold \( \eta \) is chosen not to be smaller than \( \sigma_i^2 \).

From (29), the mse of the angle-domain techniques is given by

\[
mse = \frac{1}{N_d N_t N_r} \text{trace} \left\{ \mathbf{E} \left[ \mathbf{h}^H \mathbf{M} \mathbf{h}^a \mathbf{M}^H + \mathbf{h}^a \mathbf{h}^H + 2 \mathbf{M} \mathbf{h}^a \mathbf{h}^a \mathbf{M}^H \right] \right\}
\] (39)

where \( \mathbf{v}^a \) is the angle-domain noise vector. Denoting \( \hat{h}_i \) and \( \hat{v}_i \) as the instantaneous power of the \( i \)th element of \( \mathbf{h}^a \) and \( \mathbf{v}^a \), respectively, we rewrite (39) as

\[
mse = \frac{1}{N_d N_t N_r} \sum_{i=1}^{N_d N_t N_r} \mathbf{E} \left[ (m_i - 1)^2 \hat{h}_i + m_i^2 \hat{v}_i \right].
\] (40)

More specifically

\[
\mathbf{E} \left[ (m_i - 1)^2 \hat{h}_i \right] = \int_0^\infty \hat{h}_i \mathbf{P}_h(\hat{h}_i) \int_0^\infty (m_i - 1)^2 \mathbf{P}_{r|h}(\hat{r}_i | \hat{h}_i) d\hat{r}_i d\hat{h}_i \\
+ \int_0^\infty \hat{h}_i \mathbf{P}_h(\hat{h}_i) \int_0^\eta \mathbf{P}_{r|h}(\hat{r}_i | \hat{h}_i) d\hat{r}_i d\hat{h}_i
\] (41)

is the mse part due to the filtered instantaneous power of the \( i \)th channel coefficient, and

\[
\mathbf{E} \left[ m_i^2 \hat{v}_i \right] = \int_0^\infty \hat{v}_i \mathbf{P}_v(\hat{v}_i) \int_0^\infty m_i^2 \mathbf{P}_{r|v}(\hat{r}_i | \hat{v}_i) d\hat{r}_i d\hat{v}_i
\] (42)

is the mse part due to the unfiltered noise components in the \( i \)th estimated channel coefficient, where \( \mathbf{P}_h(\hat{h}_i) \), \( \mathbf{P}_v(\hat{v}_i) \), \( \mathbf{P}_{r|h}(\hat{r}_i | \hat{h}_i) \), and \( \mathbf{P}_{r|v}(\hat{r}_i | \hat{v}_i) \) are the probability density functions of \( \hat{h}_i \) and \( \hat{v}_i \), and the conditional probability density functions of \( \hat{r}_i \) conditioned on \( \hat{h}_i \) and \( \hat{v}_i \), respectively. The channel and the noise have complex normal distributions. Thus, \( \mathbf{P}_h(\hat{h}_i) \) and \( \mathbf{P}_v(\hat{v}_i) \) have exponential distributions, and \( \mathbf{P}_{r|h}(\hat{r}_i | \hat{h}_i) \) and \( \mathbf{P}_{r|v}(\hat{r}_i | \hat{v}_i) \) have noncentral chi-square distributions [25].

Let \( \sigma_i^2 \) denote the channel power of the \( i \)th element of \( \mathbf{h}^a \). When \( \sigma_i^2 \) is not equal to zero, (41) and (42) become

\[
\mathbf{E} \left[ (m_i - 1)^2 \hat{h}_i \right] = \int_0^\infty \frac{\hat{h}_i}{\sigma_i^2} \int_0^\infty (m_i - 1)^2 \frac{e^{-\frac{\hat{r}_i + \hat{h}_i}{\sigma_f^2}}}{\sigma_f^2} d\hat{r}_i d\hat{h}_i \\
\times J_0 \left( \frac{2\sqrt{\hat{r}_i \hat{h}_i}}{\sigma_f^2} \right) d\hat{r}_i d\hat{h}_i
\] (43)

\[
\mathbf{E} \left[ m_i^2 \hat{v}_i \right] = \int_0^\eta \hat{v}_i \frac{\hat{v}_i}{\sigma_f^2} \int_0^\infty m_i^2 \frac{e^{-\frac{\hat{r}_i + \hat{v}_i}{\sigma_f^2}}}{\sigma_f^2} d\hat{r}_i d\hat{v}_i \\
\times J_0 \left( \frac{2\sqrt{\hat{r}_i \hat{v}_i}}{\sigma_f^2} \right) d\hat{r}_i d\hat{v}_i
\] (44)

respectively, where \( J_0(z) \) is the modified Bessel function of the first kind.

When \( \sigma_i^2 \) is equal to zero, \( \hat{h}_i = 0 \), and \( \hat{r}_i = \hat{v}_i \), then \( \mathbf{E} \left[ (m_i - 1)^2 \hat{h}_i \right] = 0 \), and the mse of the \( i \)th estimated channel coefficient is simplified as

\[
mse_i = \mathbf{E} \left[ m_i^2 \hat{v}_i \right] = \int_0^\eta \frac{\hat{v}_i}{\sigma_f^2} \frac{\hat{v}_i}{\sigma_f^2} d\hat{v}_i.
\] (45)

A. Performance of the MST Selection Techniques

In this subsection, we analyze the performance of the MST selection techniques and show that the optimal strategy to minimize the mse for the MST selection techniques is to ignore all the array–frequency domain LS-estimated channel coefficients, whose corresponding channel powers are smaller than the noise variance, and retain all the remaining estimated channel coefficients.

1) When \( \sigma_i^2 \) Is Not Equal to Zero: From (43) and (44), the mse of the \( i \)th estimated channel coefficient is given by

\[
mse_i = \mathbf{E} \left[ (m_i - 1)^2 \hat{h}_i \right] + \mathbf{E} \left[ m_i^2 \hat{v}_i \right]
\] (46)

Authorized licensed use limited to: Eindhoven University of Technology. Downloaded on December 11, 2008 at 06:56 from IEEE Xplore. Restrictions apply.
Differentiating $\text{mse}_i$ with respect to $\eta$, we obtain
\[ \frac{\partial \text{mse}_i}{\partial \eta} = \left( \sigma^2_i - \sigma^2_f \right) e^{-\frac{n}{\sigma^2_i + \sigma^2_f}} \frac{n}{\left( \sigma^2_i + \sigma^2_f \right)^2}. \] (47)

From (47), we find that although $\text{mse}_i$ is difficult to analytically calculate, its gradient is surprisingly simple.

- When $\sigma^2_i = \sigma^2_f$, the last two terms in (46) become identical. This is reasonable since setting the estimated channel coefficient as zero or retaining the original coarse estimate yields the same $\text{mse}_i$. Therefore, $\text{mse}_i$ will always be $\sigma^2_f$ no matter what threshold is chosen.
- When $\sigma^2_i > \sigma^2_f$, (47) and the first derivative test [25], we conclude that $\text{mse}_i$ reaches its extremum when $\eta = 0$ or $\eta = \infty$. Since (47) is always positive, $\text{mse}_i$ is monotonically increasing and has its minimum $\sigma^2_i$ at $\eta = 0$. This result indicates that the LS-estimated channel coefficient should be retained when the corresponding channel power is larger than the noise variance. It is reasonable since when we ignore the estimated channel coefficient, the resulting performance loss due to the ignorance of instantaneous power will be larger than the performance gain due to the removal of the noise.
- When $\sigma^2_i < \sigma^2_f$, (47) is always negative. Thus, $\text{mse}_i$ reaches its minimum $\sigma^2_i$ at $\eta = \infty$. This result indicates that the LS-estimated channel coefficient is required to be ignored when the corresponding channel power is smaller than the noise variance. It is reasonable as the performance gain surpasses the performance loss when the corresponding estimated coefficient is ignored.

2) When $\sigma^2_i$ Is Equal to Zero: In this case, (45) becomes
\[ \text{mse}_i = \int_{\eta=0}^{\eta=\infty} \frac{\sigma^2_f}{\sigma^2_f} e^{\frac{n}{\sigma^2_f}} d\hat{v}_i = e^{-\frac{n}{\sigma^2_f}} \left( \eta + \sigma^2_f \right). \] (48)

Differentiating $\text{mse}_i$ with respect to $\eta$, we obtain
\[ \frac{\partial \text{mse}_i}{\partial \eta} = -\eta \frac{\sigma^2_f}{\sigma^2_f} \leq 0. \] (49)

Similar to the discussion above, we find that $\text{mse}_i$ reaches its minimum at $\eta = \infty$. This result indicates that the LS-estimated channel coefficient is required to be ignored when the corresponding channel power is smaller than the noise variance. It is reasonable since the corresponding estimated channel coefficient only contains the noise, and ignoring this estimated coefficient will result in no performance loss.

From the discussion above, we conclude that the choice of threshold $\eta$ is dependent on $\sigma^2_i$ and $\sigma^2_f$, as shown in Table I, because we need to balance the performance gain and loss when we ignore the estimated channel coefficients. Thus, the optimum threshold should be obtained for each channel coefficient. Since the channel is independent of noise, the average power of the LS-estimated channel coefficient is the sum of the channel power and the noise variance (i.e., $\sigma^2_i + \sigma^2_f$). For this reason, we may set the threshold $\eta$ to be $2\sigma^2_f$. When the average power of the $i$th LS-estimated channel coefficient exceeds this threshold $2\sigma^2_f$, it means that $\sigma^2_i > \sigma^2_f$. Therefore, we set $m_i$ to be 1 [see (36)] to retain the $i$th LS-estimated channel coefficient. This is optimum to minimize the $\text{mse}_i$, as we discussed above. Similarly, this threshold is optimum to the minimization of $\text{mse}_i$, for the case when $\sigma^2_i < \sigma^2_f$. Therefore, when the average power of the LS-estimated channel coefficient is available, we can still obtain the corresponding optimal threshold.

As $\sigma^2_i$ is assumed to be not available in this paper, the average power of the corresponding LS-estimated channel coefficient may also not be available. Then, we may use the channel instantaneous power (i.e., the instantaneous power of the estimated channel coefficient) to approximate the average power of the LS-estimated channel coefficient. Due to the monotone property of $\text{mse}_i$ with the increase in $\eta$, Table I implies that for the given threshold $\eta = 2\sigma^2_f$, $\text{mse}_i$ is always smaller than $\sigma^2_f$ when $\sigma^2_i < \sigma^2_f$, and larger than $\sigma^2_f$ when $\sigma^2_i > \sigma^2_f$. Therefore, the overall performance of the MST selection techniques is dependent on the portion of the number of channel coefficients, whose average powers are below $\sigma^2_f$, to the total number of channel coefficients. When a majority of channel coefficients have average powers that are below $\sigma^2_f$ (such as in the angle–frequency domain), the MST selection technique can improve over the array–frequency domain LS technique. This is also verified in our simulation results presented in Section VI.

### B. Performance of the AMMSE Technique

As $m_i$ is dependent on the channel instantaneous power, directly analyzing the performance of the AMMSE technique results in high computational complexity. Therefore, in this subsection, we only compare the AMMSE technique with the MST selection technique to provide some general understanding on the performance trends.

1) When $\sigma^2_i$ Is Not Equal to Zero: As shown in the Appendix, the AMMSE technique performs better than the MST selection technique when $\sigma^2_i$ is not equal to zero.

### Table I

<table>
<thead>
<tr>
<th>Maximum and Minimum MSE, and the Corresponding Thresholds for the MST Selection Techniques</th>
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</thead>
<tbody>
<tr>
<td>$\sigma^2_i &gt; \sigma^2_f$</td>
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<tr>
<td>Maximum $\text{MSE}_i$</td>
</tr>
<tr>
<td>Minimum $\text{MSE}_i$</td>
</tr>
<tr>
<td>Threshold for Minimum $\text{MSE}_i$</td>
</tr>
</tbody>
</table>

5 For $\sigma^2_i = \sigma^2_f$, $\partial\text{mse}_i/\partial\eta$ is always zero, hence, indicating that $\text{mse}_i$ is constant, as claimed earlier.
2) When $\sigma^2$ Is Equal to Zero: In this case, (45) becomes

$$\text{mse}_i = \int_{\eta} m_i^2 \frac{\hat{v}_i}{\sigma^2_f} e^{-\frac{\hat{v}_i}{\sigma^2_f}} d\hat{v}_i$$

because $m_i$ is always smaller than one. This implies that the AMMSE technique performs better than the MST selection technique when $\sigma^2$ is equal to zero.

Therefore, we conclude that the AMMSE technique always performs better than the MST selection technique. Since the angle–time domain MST selection technique yields better performance than the array–frequency domain LS technique when the threshold is $2\sigma^2$, as discussed in the previous subsection, the angle–time domain AMMSE technique should achieve further performance gain.

VI. SIMULATION RESULTS

Computer simulations are carried out for the IEEE 802.11n channel models [24], [34]. As similar conclusions can be drawn from both the line-of-sight and NLOS scenarios based on our simulations, we only present the results for the NLOS scenario. We evaluated different channel estimation techniques for the five IEEE 802.11n channel models that represent various indoor environments such as residential homes and small offices. Model A corresponds to the MIMO flat-fading channel with one single cluster. This simple model serves as the basis to investigate the characteristics of cluster-based channel models in the angle domain. Here, $\{\text{AoD}_m, \text{AS}_t\}$ and $\{\text{AoA}_m, \text{AS}_r\}$ refer to the mean angle of clusters and angular spread of clusters, respectively, for the transmit and receive antennas, respectively, as illustrated in Fig. 3. All these four parameters are physically determined for a given propagation scenario and will affect the relative average power for each angle−time domain beam. In the simulation, we assign different values to these parameters for the model A to represent various propagation environments. This model is usually used for stressing the detection performance. It occurs only a small percentage of time in reality for the systems under consideration [24]. Models B–E correspond to 15-, 30-, 50-, and 100-ns root-mean-square delay spread, respectively. All these four models represent nonsample-spaced channels. As we found that the performance trends in different channel estimation techniques are similar in all these four models, we only present the results for models B and E because of their minimum and maximum degrees of freedom, respectively. In the simulations, we assign the values indicated in [24, App. C] to $\{\text{AoD}_m, \text{AS}_t\}$ and $\{\text{AoA}_m, \text{AS}_r\}$ to represent typical small environments. We also evaluated different channel estimation techniques for the ideally assumed channel model that is spatially uncorrelated and sample spaced [4], [9], [10]. This channel is set to have the exponential power delay profile with the channel length $N_h$ being 12. Throughout the simulations, we assume that $N_d = 64$ and $N_t = N_r = 4$, and that the normalized separation between adjacent antennas $\Delta_r = \Delta_t = 0.5$. Furthermore, we assume that the channel power for each link between one transmit and one receive antenna is normalized to one throughout the simulations.

A. Channel Model A

As discussed, we assign various values to $\{\text{AoD}_m, \text{AS}_t\}$ and $\{\text{AoA}_m, \text{AS}_r\}$ in channel model A to investigate the performance of the angle-domain channel estimation techniques in the presence of a single cluster. As the angular spread is usually not smaller than 40° in indoor environments [24], we consider 40° for both $\text{AS}_t$ and $\text{AS}_r$ as the worst case in performing angle-domain channel estimations. We also consider the angular spread to be 2°, which is valid for outdoor environments, for both $\text{AS}_t$ and $\text{AS}_r$ as the best case. Furthermore, the energy of multipath components from 45° leaks into more than one angle–time domain beam, which is undesirable in the angle-domain channel estimations. Thus, we consider 45° to be the $\text{AoD}_m$ and the $\text{AoA}_m$ of the cluster for the worse-case consideration. We also consider 0°, from which most of the energy of multipath components concentrates on one angle–time domain beam, to be the $\text{AoD}_m$ and the $\text{AoA}_m$. In the following figures, which represent the channel power for each angle–time domain beam (see Figs. 4 and 6), the areas of square and circle are proportional to the angle- and array-domain normalized average power, respectively, with respect to the corresponding maximum average power in the angle domain.

Fig. 4 corresponds to the best case where the angular spread is 2°. As expected, we see that the angle-domain channel power is concentrated in the $(1,1)$th angle–time domain beam, whereas the array-domain channel power is uniformly distributed over all the array–time domain beams. These observations...
Fig. 4. Normalized channel power for each angle-time domain beam of model A with AoD_m = 0°, AS_t = 2°, AoA_m = 0°, and AS_r = 2°.

Fig. 5. Performance of different channel estimation techniques for model A with AoD_m = 0°, AS_t = 2°, AoA_m = 0°, and AS_r = 2°.

indicate the effectiveness of the MST selection and AMMSE techniques. From Fig. 5, it is clear that all the angle-domain techniques improve over the array–frequency domain LS technique at all the SNRs under consideration. We also find that the angle–time domain AMMSE technique improves over the angle–time domain MST selection technique by around 3.5 dB because the former is more like an LMMSE technique. Although the MST selection is a nonlinear process, it is interesting to find that the performance of the angle-domain techniques is proportional to that of the array–frequency domain LS technique in Fig. 5 because the ignored channel instantaneous power is not significant.

We consider the worst case where the angular spread is 40° in Figs. 6 and 7. Compared to Fig. 4, the angle-domain channel power tends to leak into other angle–time domain beams. Therefore, the channel powers in all angle–time domain beams are relatively high. From Fig. 7, we find that all the angle–time domain techniques still outperform the array–frequency domain technique for all the SNRs. However, the angle–frequency domain MST selection technique does not perform well because the number of channel coefficients whose channel powers are above the noise variance is relatively large. We also observe that the angle–time domain AMMSE technique achieves the best performance, as expected.

B. Typical Channel Models

We consider typical nonsample-spaced indoor channels in Figs. 8 and 9. In such cases, AoD_m and AoA_m of clusters of multipath components tend to be uniformly distributed over all the angles. Then, the technique proposed in [23] will be the same as the array–frequency LS technique because all the angle–time domain beams are identified as signal beams. For this reason, its simulation results are not presented here. Compared with Figs. 5 and 7, we find that unlike in model A, the achieved performance gain of the angle-domain
Fig. 8. Performance of different channel estimation techniques for model B.

Fig. 9. Performance of different channel estimation techniques for model E.

Fig. 10. Performance of the angle-domain channel estimation techniques with different thresholds for model B. The number shown in the bracket indicates that the threshold is set to $\alpha \sigma_f^2$. Otherwise, the threshold is set to $2\sigma_f^2$.

techniques over the array–frequency domain LS technique is not significant at high SNRs in models B and E. This is because the ignored channel instantaneous powers are always larger in the nonsample-spaced channels. Nevertheless, the angle–time domain AMMSE technique still achieves the best performance for models B and E. Therefore, we can choose the angle–time domain AMMSE technique to perform channel estimation in the IEEE-802.11-based MIMO-OFDM systems because of its superior performance as well as its robustness.

As the channel power tends to be concentrated over a short time interval in the angle–time domain,\(^\text{7}\) the number of channel coefficients whose corresponding channel powers are below the noise variance is relatively large. For such channel coefficients, the optimum threshold should be infinite, as discussed in Section V. Intuitively, increasing the threshold will improve the estimation performance for these channel coefficients. However, this will result in an adverse effect on the estimation performance for the channel coefficients whose channel powers exceed $\sigma_f^2$. As the number of channel coefficients whose channel powers are below $\sigma_f^2$ is larger compared to that of the remaining channel coefficients, it should be interesting to investigate the overall performance of the angle–time domain techniques when the threshold is increased. Fig. 10 shows that the angle–time MST selection and AMMSE techniques are improved in terms of performance when the threshold is increased to $3\sigma_f^2$ at nearly all the SNRs. Note that the selection of threshold is a tradeoff between the performance loss due to the ignorance of channel instantaneous power and the performance gain due to the removal of noise. Therefore, we find that further increasing the threshold to $6\sigma_f^2$ degrades the performance at high SNRs because the performance loss will dominate. Note that when the threshold is relatively high, we only reserve the channel coefficients whose corresponding $||\tilde{c}_{r_i,t_i}(l)||^2 - \sigma_f^2||/||\tilde{c}_{r_i,t_i}(l)||^2$ shown in (28) approaches one. This implies that the performance of the angle–time domain MST selection technique should gradually approach that of the angle–time domain AMMSE technique with the increase in thresholds. This implication is verified in Fig. 10. Therefore, we may use the angle–time domain MST selection technique when the threshold is relatively high for typical MIMO-OFDM systems.

\(^\text{7}\)The time intervals for different angle-time domain beams are not likely to be the same because the first paths falling in each beam may asynchronously arrive.

Note that the noise variance $\sigma_f^2$ is required to be priorly known to decide the threshold in the above simulations. However, the exact knowledge of $\sigma_f^2$ may not always be available. Therefore, for a robust estimator design, we should fix the threshold for a target range of SNRs. As illustrated in Fig. 10, increasing the threshold will even improve the performance of angle–time domain techniques particularly at low SNRs. Therefore, for a given target SNR range, we may use $2\sigma_f^2$ at the lowest SNR (because of the largest $\sigma_f^2$) as the fixed threshold. In Fig. 11, we divide the whole SNR range into five disjoint groups. Each corresponds to one target SNR range, within which the threshold is fixed. Fig. 10 indicates that the threshold should be smaller (or at most slightly larger)
than the corresponding $6\sigma_f^2$ at low SNRs and smaller than the corresponding $3\sigma_f^2$ at high SNRs. These results imply that we could choose a larger factor of $\sigma_f^2$ (and, thus, more elements for each group) at low SNRs than at high SNRs. From Fig. 11, we observe that the performance of angle–time domain estimations is always better than that of the array–frequency domain LS technique. Furthermore, we find that the performance of the angle–time domain MST selection technique is close to that of the angle–time domain AMMSE technique at the highest SNR in each target range (see the points when SNR = 10, 15, 19, 22, and 25 dB in Fig. 11). These results are consistent with the observations in Fig. 10, which shows that increasing the threshold makes the performance of the angle–time domain MST selection technique approach that of the angle–time domain AMMSE technique. However, the angle–frequency domain MST selection technique may not perform well at the highest SNR for each target range because of the relatively large performance loss due to the channel instantaneous power ignorance. As the angle–time domain AMMSE technique performs best at all the SNRs under consideration, we may conclude that the angle–time domain AMMSE technique is suitable for the typical IEEE 802.11n MIMO-OFDM systems when the target range of SNRs is available.

Note that at higher SNRs, the threshold $\eta$ approaches zero, and thus, almost all the channel coefficients will not be filtered. Therefore, the angle–time domain MST selection and angle–frequency domain MST selection techniques will perform more similarly to the array–frequency domain LS technique as the SNR increases. In addition, the angle–time AMMSE technique will also perform more similarly to the angle–time domain MST selection technique as the SNR increases because the multiplication factor shown in (28) approaches to one for each angle–time domain channel coefficient. In summary, all the channel estimation techniques will converge at a relatively high SNR, as implied in all the figures shown in this subsection. Thus, the angle–time domain AMMSE technique is more preferable at lower SNRs than at higher SNRs because of the larger achievable performance gain.

C. Spatially Uncorrelated Channel

In Sections VI-A and B, we consider IEEE 802.11n channel models that are spatially correlated and show that the angle-domain channel estimation techniques can achieve good performance. In this subsection, we consider the sample-spaced spatially uncorrelated channel. In such a channel, the angle–time domain and array–time domain channel coefficients have the same statistics. Still, as the channel coefficients in the angle–time domain are uncorrelated with each other, from Fig. 12, we find that the AMMSE technique in this domain can achieve the best performance. Therefore, compared with the array–frequency domain LS technique, we can always use the angle-domain AMMSE technique for different channels to achieve higher performance at the price of higher complexity.

VII. Conclusion

In this paper, the angle-domain representation is used to describe MIMO-OFDM systems. Based on this representation, we propose the angle–frequency domain MST selection technique, the angle–time domain MST selection technique, and the angle–time domain AMMSE technique. These three techniques do not require prior knowledge of the channel correlation and are shown to be effective when the angular spread of clusters of multipath components is small. More importantly, the angle–time domain techniques can improve over the array–frequency domain LS technique in all the cases under consideration even when the angular spread is large, at the price of higher complexity. Furthermore, our theoretical analysis and simulation results indicate that the angle–time domain AMMSE technique achieves the best performance and is robust to the choice of threshold and mismatch of operating an SNR. Thus, when only the target SNR range is available to the receiver,
the angle–time domain AMMSE technique is a potential candidate for typical IEEE 802.11n MIMO-OFDM systems. In addition, we also find that with a suitable threshold and a known operating SNR, the angle–time domain MST selection technique results in little performance degradation compared to the angle–time domain AMMSE technique. Therefore, in such cases, the angle–time domain MST selection technique may be used for IEEE 802.11n MIMO-OFDM systems because of its lower computational complexity compared to the angle–time domain AMMSE technique.

**Appendix**

**Performance Comparison of AMMSE and MST Selection Techniques**

In this appendix, we analytically compare the performance of the AMMSE and MST selection techniques. From (43) and (44), the mse due to the $i$th estimated channel coefficient is given by

$$
\text{mse}_i = \int_0^\infty \frac{\hat{v}_i}{\hat{\sigma}_i^2} e^{-\frac{\hat{v}_i}{\hat{\sigma}_i^2}} \int_0^\infty (\hat{r}_i - \hat{\sigma}_i^2)^2 e^{-\frac{\hat{r}_i}{\hat{\sigma}_i^2}} d\hat{r}_i d\hat{\sigma}_i + \int_0^\infty \frac{\hat{h}_i}{\hat{\sigma}_i^2} e^{-\frac{\hat{h}_i}{\hat{\sigma}_i^2}} \int_0^\infty (\hat{r}_i - \hat{\sigma}_i^2)^2 e^{-\frac{\hat{r}_i}{\hat{\sigma}_i^2}} d\hat{r}_i d\hat{\sigma}_i \\
\times J_0 \left( \frac{2\sqrt{\hat{r}_i \hat{h}_i}}{\hat{\sigma}_i^2} \right) d\hat{r}_i d\hat{h}_i \\
+ \int_0^\infty \frac{\hat{h}_i}{\hat{\sigma}_i^2} e^{-\frac{\hat{h}_i}{\hat{\sigma}_i^2}} \int_0^\infty \frac{\hat{r}_i + \hat{h}_i}{\hat{\sigma}_i^2} \left( \frac{2\sqrt{\hat{r}_i \hat{h}_i}}{\hat{\sigma}_i^2} \right) d\hat{r}_i d\hat{h}_i.
$$

(51)

Compared to (46), the difference of mse$_i$, which is contributed by the $i$th channel coefficient, between the AMMSE technique and the MST selection technique is given by

$$
\text{DIF}_i = \int_0^\infty \frac{\hat{v}_i}{\hat{\sigma}_i^2} e^{-\frac{\hat{v}_i}{\hat{\sigma}_i^2}} \int_0^\infty \frac{\hat{r}_i + \hat{v}_i}{\hat{\sigma}_i^2} d\hat{r}_i d\hat{\sigma}_i \\
\times J_0 \left( \frac{2\sqrt{\hat{r}_i \hat{v}_i}}{\hat{\sigma}_i^2} \right) d\hat{r}_i d\hat{\sigma}_i + \int_0^\infty \frac{\hat{h}_i}{\hat{\sigma}_i^2} e^{-\frac{\hat{h}_i}{\hat{\sigma}_i^2}} \int_0^\infty \frac{\hat{r}_i + \hat{h}_i}{\hat{\sigma}_i^2} \left( \frac{2\sqrt{\hat{r}_i \hat{h}_i}}{\hat{\sigma}_i^2} \right) d\hat{r}_i d\hat{h}_i \\
- \frac{2}{\hat{\sigma}_i^2} \int_0^\infty \frac{\hat{v}_i}{\hat{\sigma}_i^2} e^{-\frac{\hat{v}_i}{\hat{\sigma}_i^2}} \int_0^\infty \frac{\hat{r}_i + \hat{v}_i}{\hat{\sigma}_i^2} d\hat{r}_i d\hat{\sigma}_i \\
\times J_0 \left( \frac{2\sqrt{\hat{r}_i \hat{v}_i}}{\hat{\sigma}_i^2} \right) d\hat{r}_i d\hat{\sigma}_i.
$$

(52)

Differentiating DIF$_i$ with respect to $\eta$, we obtain

$$
\frac{\partial \text{DIF}_i}{\partial \eta} = \frac{\sigma_i^2 e^{-\frac{\eta}{\sigma_i^2}} f(\sigma_i^2, \sigma_i^2, \eta)}{(\sigma_i^4 + \sigma_i^2 \eta)^\frac{3}{2}}
$$

(53)

where

$$
f(\sigma_i^2, \sigma_i^2, \eta) = 2\sigma_i^2 \eta^2 + (\sigma_i^4 + 2\sigma_i^2 \sigma_i^2 - \sigma_i^4) \eta - 2\sigma_i^2 \sigma_i^2 (\sigma_i^2 + \sigma_i^2)$$

$$
= 2\sigma_i^2 \left( \eta - \frac{\sigma_i^4 - 2\sigma_i^2 \sigma_i^2 - \sigma_i^4}{4\sigma_i^2} \right)^2 - 2\sigma_i^2 \sigma_i^2 (\sigma_i^2 + \sigma_i^2)$$

$$
- \frac{(\sigma_i^4 - 2\sigma_i^2 \sigma_i^2 - \sigma_i^4)^2}{8\sigma_i^2}.
$$

(54)

As the threshold $\eta$ is not smaller than $\sigma_i^2$, which is larger than $(\sigma_i^4 - 2\sigma_i^2 \sigma_i^2 - \sigma_i^4)/(4\sigma_i^2)$, (54) reaches its minimum 0 when the threshold $\eta$ is equal to $\sigma_i^2$. Therefore, when the threshold $\eta$ is larger than $\sigma_i^2$, (54) is larger than 0 (i.e., positive), and so is (53). Therefore, DIF$_i$ is monotonically increasing with the increase in $\eta$ and reaches its maximum 0 when $\eta = \infty$. Consequently, DIF$_i$ is always negative, which implies that the AMMSE technique performs better than the MST selection technique when $\sigma_i^2$ is not equal to zero.

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