Phase Noise Compensation for OFDM Systems
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Abstract—We describe a low complexity method for time domain compensation of phase noise in OFDM systems. We extend existing methods in several respects. First we suggest using the Karhunen-Loève representation of the phase noise process to estimate the phase noise. We then derive an improved data-directed choice of basis elements for LS phase noise estimation and present its total least square counterpart problem. The proposed method helps overcome one of the major weaknesses of OFDM systems. We also generalize the time domain phase noise compensation to the multiuser MIMO context. Finally we present simulation results using both simulated and measured phase noise. We quantify the tracking performance in the presence of residual carrier offset.

I. INTRODUCTION

OFDM has become a prominent technology that is utilized in many modern communication systems including cellular systems such as 3GPP LTE [2], wireless LAN (WLAN) [3] and WiMax [4]. In future LTE systems (see 802.11ac) multicarrier modulations such as Generalized frequency division multiplex (GFDM), Filter bank multi-carrier (FBMC), Universal filtered multi-carrier (UFMC) and Filtered OFDM (f-OFDM) [5]–[8] may be implemented. In spite of its popularity and robustness to multipath propagation, OFDM is known to be extremely affected by phase noise (PN) and frequency offset [9]–[13]. MIMO OFDM receivers are very sensitive to the phase noise coming from the difference between the carrier frequency and the local oscillator (LO). In the case of very high data rates this is actually the limiting factor on performance. High order modulations such as 256 QAM are also severely affected by phase noise. Phase noise is typically modeled as a multiplicative noise. When the LO is locked to the carrier frequency, the phase noise is lower and is modeled as a finite power random process. When the LO is not locked the phase noise is modeled as a Wiener process with infinite power, the effect of such phase noise is analyzed from an information theoretic perspective in [14]. In all typical 802.11 and LTE implementations the LO is locked; hence, we will concentrate on the first model. Phase noise can be considered to have two components: a common phase error (CPE) that is common to all carriers, and a time varying part that is frequency dependent. This part is typically weaker than the CPE and generates the undesirable and harmful ICI.

A popular approach to ICI mitigation is the use of an MMSE equalizer in the frequency domain that balances the AWGN and the colored ICI. This approach can be found in many articles including [13], [15]–[18]. The alternative approach of PN cancellation/mitigation aims at jointly detecting the data and cancelling the PN as described in [19], [20], or jointly estimating the channel and PN [21]–[25] and even jointly estimating the channel, detecting the data and compensating for the PN as in [26]–[29]. These schemes suffer from high complexity that is impractical to implement at high rates and in high spectral efficiency communication systems. Yet a third approach to reducing ICI is time domain processing. Casas et al. [30] proposed a LS approach in the time domain where they represented the noise using one of two fixed bases: DFT or DCT. The dominant phase noise components are then estimated using LS fitting of a few basis vectors (typically the low frequency components due to phase noise properties). When a single basis vector is used, the method reduces to that of [15]–[17]. The main drawback of the method in [30] is that typically phase noise cannot be compactly represented in the fixed basis. Another LS PN compensation scheme estimates both the channel coefficients and the PN with assumed low complexity; however, the assumption that the number of pilot subcarriers is larger than the number of transmitting antennas leads to high complexity in massive MIMO communication systems [31]–[33]. Another issue we consider is time variation of the statistical properties of the PN process. Even when the LO is locked, the statistical properties of the PN are time varying due to external conditions such as temperature; e.g., heat from the mobile device, thus making real time basis selection very desirable for practical implementations. Therefore, statistical knowledge of the PN covariance matrix should be acquired.

Phase noise is present in many communication systems such as WLAN, millimeter wave systems, full-duplex systems and massive MIMO systems. Full-duplex systems are especially affected by phase noise since self-interference is performed in order to extract the received signals, see for example [34]–[36]. The phase noise prevents transmitters from properly subtract their self-interference, this is especially harmful in high order modulations which demand high values of received SNR which are translated to low values of error vector magnitudes (EVM). OFDM millimeter wave systems are also affected by phase noise, see for example [37], [38]; here, as in other systems the effect of phase noise is most severe when using high order modulations, and is a key issue in implementing these systems. Phase noise is also present in optical communication systems using coherent optical OFDM [39]. In such networks the phase is estimated digitally without using a optical phase-locked loop, however, this estimation is not perfect and thus phase noise is cause this imperfect estimation. The phase compensation scheme that we present is adaptable to the transmission of OFDM symbols in all of these systems.
In this paper we replace the fixed basis proposed in previous works with an adaptive basis which is the best representation of the noise with respect to the $L_2$ norm for a random noise process [1]. Since in locked systems the phase noise behavior is quite stationary we can either pre-calibrate the phase noise PSD and then use an eigen-decomposition of the covariance matrix or estimate in real time the basis elements as well as the LS coefficients of the phase noise. The latter is more robust to environmental changes such as temperature which might affect the statistical properties of the phase noise. We can also replace the LS estimation of the coefficient with a total least square estimation [40] that considers imperfection of the model.

The main contributions of this article are: 1) Utilizing the Karhunen-Lo`eve representation of the phase noise process covariance matrix as basis elements. This dramatically improves the results of [30]. 2) The introduction of the implementation of the total least square estimator for phase noise mitigation schemes. 3) We efficiently track of the subspace of the covariance matrix of the phase noise process for system with no information regarding the covariance matrix of the phase noise process. This is performed utilizing the PAST algorithm [41] and behaves well even in the presence of carrier frequency offset. 4) We extended the above contributions for multiuser uplink beamforming OFDM systems.

This paper is organized as follows: in section II we present the phase noise model. Section III presents our results on phase noise compensation in SISO systems and discusses the computational aspects of the compensation scheme. In Section IV we discuss two enhancements to the compensation method. Section V is dedicated to tracking the dominant subspace of the Karhunen-Lo`eve (KL) representation presented in Section III. In Section VI we present simulation and measured results of the proposed phase noise compensation method. Section VI-A covers simulated phase noise whereas Section VI-B discusses measured phase noise. Section VI-C analyzes simulations of the tracking algorithm proposed in Section V for the measured phase noise. Finally, Section VII concludes the paper.

Notations: We denote the convolution between two continuous time signals $x$ and $y$ at time $t$ by $x * y(t)$; for discrete time we denote the convolution between the two discrete time signals $x$ and $y$ at time $k$ by $x * y(k)$. Vectors and matrices appear in bold. Let $a$ be a vector, we denote by $a^T$ the transpose vector of $a$ and by $a^*$ the conjugate transpose of $a$; note that if $a$ is of dimension 1; i.e., scalar, then $a^*$ is equal to the conjugate of $a$. Moreover, $M^*$ is the Moore-Penrose pseudoinverse of a matrix $M$. Finally, $|| \cdot ||$ denotes the $L_2$ norm and $|| \cdot ||_F$ denotes the Frobenius norm.

II. PHASE NOISE MODEL

In this section we describe a mathematical model for the phase noise process and its effects on OFDM systems. Consider an OFDM system described by

$$x(t) = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} s(k)e^{j\omega_k t}, \quad 0 \leq t \leq T_s$$

where $\omega_k = \omega_0 + k\Delta\omega$ is the frequency of the $k$’th channel, $k = -\frac{N}{2}, \ldots, \frac{N}{2}$, $\omega_0$ is the carrier frequency and

$$\Delta\omega = \frac{2\pi}{T_s}$$

is the angular sampling frequency. Additionally, $s(k)$ is the symbol transmitted of the $k$’th channel and is independent of symbols transmitted over other channels. The OFDM symbol passes through a time invariant channel (we assume a quasi-stationary fading process) and the received signal $y(t)$ is given by

$$y(t) = h * x(t) + n(t).$$

Phase noise is multiplicative noise resulting from the jitter of the LO of the OFDM system. We can model the received signal with the effect of the phase noise as

$$z(t) = y(t)e^{j\phi(t)},$$

where $\phi(t)$ is a random process that can be considered to be a filtered Gaussian process with PSD $P_\phi(f)$. The process $\psi(t) = e^{j\phi(t)}$ is the multiplicative noise process that can also include the residual frequency offset (a linear phase component) and the common phase error that is constant across frequencies. We want to estimate this and remove its effect, since it introduces inter-channel interference (ICI).

We assume that $\psi(t)$ is a stationary process with a known covariance $r_\psi(\tau) = E[\psi(t)\psi^*(t-\tau)]$. This assumption is very reasonable when the LO is locked to a stable frequency source through a Phase-Locked Loop (PLL).

Let $\psi = (\psi_1, \ldots, \psi_N)^T$ be a vector of $N$ consecutive samples of the phase noise process $\psi_m = \psi(mT_s)$. We define the covariance matrix of the phase noise process by

$$R_{\psi\psi} = \begin{bmatrix} E(\psi_1\psi_1^*) & \cdots & E(\psi_1\psi_N^*) \\ \vdots & \ddots & \vdots \\ E(\psi_N\psi_1^*) & \cdots & E(\psi_N\psi_N^*) \end{bmatrix}.$$  

When trying to represent the phase noise along a single OFDM symbol it is natural to use the basis of the eigenvectors of $R_{\psi\psi}$. We decompose $R_{\psi\psi}$ using an eigen-decomposition as

$$R_{\psi\psi} = \sum_{i=0}^{N-1} \mu_i u_i u_i^*,$$

where $\mu_0, \ldots, \mu_{N-1}$ denote the eigenvalues of $R_{\psi\psi}$ and $u_0, \ldots, u_{N-1}$ denote their respective eigenvectors.

We now describe the received signal and channel. We begin with a SISO model and then extend it to a SIMO model. Note that we are only interested in the SIMO case since the phase noise is identical on all spatial channels. We assume that the OFDM symbols are synchronized and that the cyclic prefix
has been removed, so that the channel matrix can be assumed to be circulant, and thus be given by

\[ H = \begin{bmatrix}
    h_0 & h_1 & \cdots & \cdots & h_{N-1} \\
    h_{N-1} & h_0 & \cdots & \cdots & h_{N-2} \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    h_1 & \cdots & \cdots & h_{N-1} & h_0 
\end{bmatrix}. \tag{7} \]

Furthermore, we consider a single OFDM symbol. The time domain OFDM symbol is given by

\[ x = F_N s \tag{8} \]

where

\[ s = [s_0, \ldots, s_{N-1}]^T \tag{9} \]

is the frequency domain OFDM symbol, and

\[ F_N = \frac{1}{\sqrt{N}} \begin{bmatrix}
    1 & 1 & \cdots & \cdots & 1 \\
    1 & e^{-2\pi i 1/N} & \cdots & \cdots & e^{-2\pi i (N-1)/N} \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    1 & e^{-2\pi i (N-1)/N} & \cdots & \cdots & e^{-2\pi i (N-1)(N-1)/N} 
\end{bmatrix}. \tag{10} \]

is the DFT matrix. Let \( n = [n_0, \ldots, n_{N-1}] \) be the additive white Gaussian noise. It follows that

\[ y = HF_N^* s + n = Hx + n \tag{11} \]

is the received signal when no phase noise is present.

The received OFDM symbol \( z = [z_0, \ldots, z_{N-1}]^T \) is given by

\[ z = e^{j\Phi} (Hx + n) \tag{12} \]

where

\[ \Phi = \text{diag}(\phi_0, \ldots, \phi_{N-1}) \tag{13} \]

is the phase noise vector.

Define a received data matrix \( Z \); this matrix is given by

\[ Z = \text{diag}(z) = \begin{bmatrix}
    z_0 & \cdots & \cdots & \cdots & \cdots \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    z_{N-1} & \cdots & \cdots & \cdots & \cdots 
\end{bmatrix} = Ye^{j\Phi} \tag{14} \]

where

\[ Y = \text{diag}(y) = \begin{bmatrix}
    y_0 & \cdots & \cdots & \cdots & \cdots \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    \vdots & \ddots & \ddots & \ddots & \vdots \\
    y_{N-1} & \cdots & \cdots & \cdots & \cdots 
\end{bmatrix}. \tag{15} \]

Our problem is to estimate the phase noise and construct a time domain vector that cancels out the harmful effect of the phase noise.

III. TIME DOMAIN COMPENSATION IN SISO SYSTEMS

We now describe a time domain method for reducing the phase noise. The idea is to use available pilot data to estimate the coefficients of a representation of the phase noise. We first present the phase compensation algorithm in [30]. Then we discuss the choice of basis for the phase noise compensation. We show that using a fixed basis such as Fourier vectors or discrete cosine transform vectors does not yield large gains in terms of ICI cancellation. We then propose the use of data directed basis selection and show the improvement achieved through this approach. This is important for ICI cancellation since common phase noise removal involves choosing the first basis vector to be the \( N \) dimensional all ones vectors \( \mathbb{I}_N = [1, \ldots, 1]^T \).

Let \( v_0, \ldots, v_{N-1} \) be a basis for \( \mathbb{C}^N \). Denote the phase noise realization by \( e^{j\Phi} = [e^{j\phi_0}, \ldots, e^{j\phi_{N-1}}]^T \) and let \( \gamma = [\gamma_0, \ldots, \gamma_{N-1}]^T \) satisfy

\[ e^{-j\Phi} = \sum_{k=0}^{N-1} \gamma_k v_k, \tag{16} \]

or equivalently

\[ e^{-j\Phi} = V\gamma, \tag{17} \]

where \( V = [v_0, \ldots, v_{N-1}] \). If we allow only \( d \) basis vectors \( V^{(d)} = [v_0, \ldots, v_{d-1}] \) we can pose the problem as a least squares problem. Had we known the vector \( e^{-j\Phi} \) our objective would have been finding \( \tilde{\gamma} \in \mathbb{C}^N \) such that \( V^{(d)}\tilde{\gamma} \) cancels the phase noise optimally (in LS sense), i.e.

\[ \tilde{\gamma} = \arg \min_{\gamma} \| e^{-j\Phi} - V^{(d)}\gamma \|^2. \tag{18} \]

However, since \( e^{-j\Phi} \) is the vector which we want to estimate, we cannot use this naive approach. We discuss ways to overcome this issue below.

A. LS compensation based on [30]

Since we do not know the phase noise \( e^{-j\Phi} \) we rely on known OFDM pilot tones. In this case we need to modify (18) assuming that we have known values \( s_p = [s_1, \ldots, s_r]^T \in \mathbb{C}^{n_{\text{pilot}}} \). Let \( \hat{y} \) be an estimate of the time domain symbol with the phase noise removed:

\[ \hat{y} = ZV^{(d)}\tilde{\gamma} \simeq Hx + n = HF_N^* s + n \tag{19} \]

Since \( H \) is diagonalized by the DFT matrix \( F_N \); i.e., \( H = F_N^* H F_N \), we obtain that

\[ \hat{s} = \Lambda^{-1} F_N ZV^{(d)}\tilde{\gamma}. \tag{20} \]

is an estimate of the received OFDM frequency domain symbol. Defining

\[ W = \Lambda^{-1} F_N ZV^{(d)}, \tag{21} \]

we obtain that our LS estimate of \( \gamma \) is given by

\[ \hat{\gamma} = \arg \min_{\gamma} \| s - W\gamma \|^2. \tag{22} \]
Therefore, we obtain that
\[ \hat{\gamma} = W_p s_p, \]  
where \( W_p \) is obtained by choosing the rows that correspond to pilot tones alone. The estimate of the phase noise cancellation vector is now given by
\[ e^{-j\varphi} = V \hat{\gamma}. \]  

Figure 1 depicts this phase noise compensation scheme.

Note that the components of \( Z \) are affected by noise and the noise is multiplied by \( \Lambda^{-1} F_N \). This suggests that the estimation of \( \gamma \) can be improved using total least squares (TLS) instead of the LS described above. We discuss this in the next section.

### B. TLS compensation

As discussed above, in the training period, the training symbols \( s_{\text{pilot}} \) are sent over the channel. Since these training symbols are predefined and known by the receiver, our compensation problem can be represented as a data least squares (DLS) problem [42]. However, since we use the basis \( V^{(d)} \) of size \( d \leq N \) we also consider discrepancies in \( s_p \).

By Eq. (20) we have that
\[ s \approx \Lambda^{-1} F_N Z V^{(d)} \hat{\gamma}. \]
The uncertainty in the model is caused by the estimation of the channel matrix \( H \), the additive noise of the channel and the reduced basis dimensions. It it therefore natural to consider the TLS estimation of \( \gamma \). Let \( \Delta W \) and \( \Delta s \) be such that
\[ (W_p + \Delta W) \gamma = s_p + \Delta s \]
where as before, \( W_p \) is obtained by choosing the rows that correspond to pilot tones alone.

The TLS problem is then
\[ \arg\min_{\gamma, \Delta W, \Delta s} \| (\Delta W, \Delta s) \|_F \quad \text{s.t.:} \quad (W_p + \Delta W) \gamma = s_p + \Delta s \]
where as stated above (Section I) the notation \( \| \cdot \|_F \) denotes the Frobenius norm. The solution in terms of \( \hat{\gamma} \) is obtained by following Algorithm 1 in [40] which we describe next. Let \( W_p \) and \( s_p \) as stated above and let
\[ [W_p, s_p] = U \Sigma Q^T \]
be the singular value decomposition (SVD) of \([W_p, s_p] \in \mathbb{C}^{n_{\text{pilot}} \times (d+1)}\). Denote
\[ Q \triangleq \begin{bmatrix} q_{11} & q_{12} \\ q_{21} & q_{22} \end{bmatrix} \]
where \( q_{11} \in \mathbb{C}^{d \times d}, q_{12} \in \mathbb{C}^{d \times 1}, q_{21} \in \mathbb{C}^{1 \times d} \) and \( q_{22} \in \mathbb{C} \). If \( q_{22} \neq 0 \), note that \( q_{22} \) is scalar, then
\[ \hat{\gamma}_{\text{TLS}} = -q_{12}/q_{22}. \]
When \( q_{22} = 0 \), there is no solution and we set \( \hat{\gamma}_{\text{TLS}} = \hat{\gamma}_{\text{LS}} \).

Note that in some practical implementations we trade off accuracy for lower complexity. However, if an SVD computation engine is available because of beamforming, for example, then this SVD engine can be used for solving the TLS problem in (28).

### C. The basis vectors

We now turn to the question of the choice of the basis vectors \( v_0, \ldots, v_{N-1} \). The authors of [30] proposed using either the columns of the DFT matrix \( F_N \) or the columns of the DCT matrix. As will be seen in the simulations, this typically leads to minor improvement over simply cancelling the common phase but the ICI is still significant. We suggest a different approach and choose the basis elements using the properties of the phase noise process. We assume that the phase noise process has covariance
\[ R_{\psi \psi} = \sum_{k=1}^{N} \mu_k u_k u_k^* \]
where \( u_0, \ldots, u_{N-1} \) are the eigenvectors corresponding to eigenvalues \( \mu_0 > \ldots > \mu_{N-1} \), respectively. This basis is the best choice for representing random realizations of a random process with covariance \( R_{\psi \psi} \) (this is basically a KL representation of the process). Since the statistical properties of the phase noise process are stationary for quite long periods they can be calibrated in advance.

### D. Computational aspects

The compensation of the phase noise involves \( O((N \log N) \times d + (n_{\text{pilot}} \times d^2) \) complex multiplications where \( d \) is the number of the basis elements. This is due to the FFT complexity adding \( d \) multiplications per symbol and the additional complexity of the LS itself. We next explain these values in more detail. Forming the matrix \( W \) involves computing
\[ W = \Lambda^{-1} F_N Z V^{(d)}. \]

Matrices \( \Lambda \) and \( Z \) are diagonal matrices and matrix \( F_n \) is symmetric. Thus, calculating of the multiplication \( ZV^{(d)} \) requires at most \( N \times d \) complex multiplication operations. Using the fast Fourier transform (FFT), we have that the complexity of calculating \( F_N Z V^{(d)} \) is \( O((N \log N) \times d) \). The additional multiplication by \( \Lambda^{-1} \) does not increase the complexity.

Solving the LS problem (22) only involves matrices of size \( n_{\text{pilot}} \times d \) resulting in \( O(n_{\text{pilot}} \times d^2) \) operations. It follows that the complexity depends on \( d \) and \( n_{\text{pilot}} \) and is \( O((N \log N) \times d + n_{\text{pilot}} \times d^2) \).

We note that the complexity of calculating the eigenvectors of the covariance matrix is \( O(N^3) \). This is done once and thus the complexity over time vanishes. However, since phase noise statistics can slowly vary due to frequency offset variations, we also provide a tracking solution. In Section V we demonstrate that we can track the \( d \) leading eigenvectors with a complexity of \( O(N d) \) operations per update. We can conclude that the complexity of our compensation scheme is \( O((N \log N) \times d + n_{\text{pilot}} \times d^2) \) even if tracking the \( d \) leading eigenvectors is performed. This is in par with the complexity of the compensation scheme presented in [30] while providing superior results due to the better and adaptive basis selection.
IV. EXTENSIONS

We now describe two extensions to the proposed method. These extensions can contribute substantially to the performance of the proposed scheme by increasing the number of available equations for PN mitigation.

A. Using null tones

While channel estimation can only use tones in which energy has been transmitted, null tones can also provide information on ISI. This depends, of course, on the amount of adjacent channel suppression; however, when the adjacent channel suppression is good, the ICI can be estimated based on the null tones as well.

B. Phase estimation in MIMO system

When a MIMO system is used, typically all transceiver chains use the same LO. Hence the phase noise can be jointly estimated based on the pilot symbols from all the receive antennas. This extra information substantially enhances the applicability of the proposed method and improves the quality of the LS fitting of the coefficients, especially in modern 802.11ac and massive MIMO systems.

Next we provide a detailed example for adapting the algorithm presented in this paper to multiuser uplink beamforming OFDM. The example assumes transmitter phase compensation [43], [44], for this reason it focuses on compensating for the receiver phase noise. Under the assumption of proper phase noise compensation at the transmitters, the phase noise at the receiver is the dominating noise compared with the compensated phase noise at the transmitters.

Suppose that a receiver with $N_r$ antennas serves $N_u$ single antenna users. Each user $k_u \in \{1, \ldots, N_u\}$ sends the vector of symbols $s_{k_u} = [s_{k_u,0}, \ldots, s_{k_u,N-1}]^T$, by aiming to transmit the signal

$$x_{k_u}(t) = \frac{1}{\sqrt{N}} \sum_{k=0}^{N-1} s_{k_u,k} e^{j\omega_k t}, \quad 0 \leq t \leq T_s. \quad (33)$$

Denote by $\phi_{k_u}(t)$ the residual phase noise at transmitter $k_u$ after performing phase noise compensation at the transmitter. The transmitted signal with the residual phase noise is given by

$$\tilde{x}_{k_u}(t) = e^{j\phi_{k_u}(t)} x_{k_u}(t) \approx x_{k_u}(t) \quad (34)$$

Denote by $h_{k_u,k_r}(t)$ the channel between user $k_u$ and the receiver antenna $k_r$. The receiving signal at antenna $k_r$ is

$$y_{k_r}(t) = \sum_{k_u=1}^{N_u} h_{k_u,k_r} \ast \tilde{x}_{k_u}(t) + n_{k_r}(t). \quad (35)$$

Assuming a common LO for all the receiving antennas (which is the case in many communication systems), we can model the received signal with the effect of the phase noise at the receiver as

$$z_{k_r}(t) = e^{j\phi(t)} y_{k_r}(t); \quad (36)$$

Note that (34) displays our assumption that adequate phase noise compensation was performed by the transmitters, thus the remaining phase noise at the transmitters is negligible.
an example of multiple LOs at the receiving end is discussed for example in [45]. Since we discuss in this example systems with transmitter phase noise compensation, we can reasonably assume that $e^{j\phi(t)}$ is considerably larger than $e^{j\phi_k(t)}$. Also, as before, we assume that the OFDM symbols are synchronized and that the cyclic prefix has been removed, so that the channel matrix can be assumed to be circulant, and thus for every antenna the channel is given by

$$H_{k_a,k_r} = \begin{bmatrix} h_{k_a,k_r,0} & h_{k_a,k_r,1} & \cdots & \cdots & h_{k_a,k_r,N-1} \\ h_{k_a,k_r,N-1} & h_{k_a,k_r,0} & \cdots & \cdots & \vdots \\ \vdots & \ddots & \ddots & \ddots & \vdots \\ \vdots & \cdots & \ddots & \ddots & \vdots \\ h_{k_a,k_r,1} & \cdots & \cdots & h_{k_a,k_r,N-1} & h_{k_a,k_r,0} \end{bmatrix}$$

(37)

Let $H_{k_r} = [H_{1,k_r}, \ldots, H_{N_r,k_r}]$ and $x = [x_1^T, \ldots, x_{N_r}^T]^T$ where

$$x_{k_u} = F_N^s k_u.$$  

(38)

Further, let

$$\tilde{x}_{k_u} = e^{j\Phi_{k_u}} x_{k_u}$$

(39)

where $\Phi_{k_u} = \text{diag}(\phi_{k_u,0}, \ldots, \phi_{k_u,N-1})$ and $\phi_{k_u,m} = \phi_{k_u}(mT_s)$. Let $\tilde{x} = [\tilde{x}_1^T, \ldots, \tilde{x}_{N_r}^T]^T$, we have that

$$y_{k_r} = \sum_{k_u=1}^{N_u} H_{k_u,k_r} \tilde{x}_{k_u} + n_{k_r} = H_{k_r} \tilde{x} + n_{k_r}.$$  

(40)

Defining $\Phi$ as in (13), we write

$$z_{k_r} = e^{j\Phi} y_{k_r} = e^{j\Phi}(H_{k_r} \tilde{x} + n_{k_r}).$$  

(41)

Now, let $H = [H_1^T, \ldots, H_{N_r}^T]^T$ and $n = [n_1^T, \ldots, n_{N_r}^T]^T$; define $y$ and $z$ as follows

$$y = \begin{bmatrix} y_1 \\ \vdots \\ y_{N_r} \end{bmatrix} = \begin{bmatrix} H_1 \tilde{x} + n_1 \\ \vdots \\ H_{N_r} \tilde{x} + n_{N_r} \end{bmatrix} = H \tilde{x} + n$$  

(42)

$$z = \begin{bmatrix} z_1 \\ \vdots \\ z_{N_r} \end{bmatrix} = \begin{bmatrix} e^{j\Phi} y_1 \\ \vdots \\ e^{j\Phi} y_{N_r} \end{bmatrix} = (I_{N_r} \otimes e^{j\Phi})y.$$  

(43)

Denote

$$Y = \text{diag}(y)$$  

(44)

it follows that

$$Z \triangleq \text{diag}(z) = Y (I_{N_r} \otimes e^{j\Phi}),$$  

(45)

where $\otimes$ is the Kronecker product\(^2\). Let $V$ be defined to be a basis of $R_{\psi\psi}$ (see (5)), and define $\gamma$ as in (16) and (17). We can conclude that

$$(I_{N_r} \otimes e^{j\Phi})(I_{N_r} \otimes (V\gamma)) = I_{N_r}.$$  

(46)

\(^2\)We note that in Figure 1 and Figure 2 the symbol $\otimes$ denotes the scalar multiplication. However, this is a special case of the Kronecker product.

Choosing a ZF beamforming matrix $B$ and assuming that $N_r > N_u$, we have that

$$s \approx (I_{N_u} \otimes F_N)BZ(1_{N_r} \otimes (V\gamma))$$  

(47)

where $1_{N_u}$ is a column vector of ones of size $N_r$. Define

$$W = (I_{N_u} \otimes F_N)BZ(I_{N_r} \otimes V)(1_{N_r} \otimes I_N).$$  

(48)

Since

$$(1_{N_r} \otimes (V\gamma)) = (I_{N_r} \otimes (V)) (1_{N_r} \otimes I_N)\gamma$$  

(49)

it follows from (47) and from our assumption that adequate phase noise compensation was performed by the transmitters

$$s \approx W\gamma.$$  

(50)

We can obtain the linear LS estimation by solving

$$\hat{\gamma} = \arg \min_{\gamma} \|s - W\gamma\|^2.$$  

(51)

Therefore, as before, we obtain that

$$\hat{\gamma} = W_p s_p,$$

(52)

where $W_p$ is obtained by choosing the rows that correspond to pilot tones alone of the different users. Alternatively, we can solve the respective TLS problem.

V. TRACKING THE DOMINANT SUBSPACE OF THE KARHUNEN-LOEVE REPRESENTATION

There are several methods to obtain the correlation matrix $R_{\psi\psi}$. The first is to pre-calibrate it and generate fixed basis vectors that are either measured or computed from the LO design. Hence, one can estimate $R_{\psi\psi}$ from the data and apply an eigen-decomposition to obtain the basis vectors. These alternatives are hard to implement since they depend on component variability in manufacturing and therefore needs to be performed for each chip separately. This might also lead to performance degradation due to environmental changes such as temperature or vendor dependent behavior. Furthermore, residual carrier offset also affects the optimal basis. A better choice that makes it possible to overcome the non-stationarity of the phase noise process is to track a basis for the phase noise subspace. To track the phase noise vectors we propose using the PAST algorithm [41]. Since we do not require our basis elements to be orthogonal we do not need the deflation approach of PAST-d (see [41]). Alternatively, the subspace can be tracked using traditional methods such that appear in [46]. The PAST algorithm is implemented as follows.

A. The PAST algorithm for subspace tracking

Let $V_{0}^{(d)}$ be a matrix composed of the $d$ low frequency vectors of the DFT matrix or any other a-priori estimate of the phase noise dominant eigenvectors (i.e., eigenvectors of $R_{\psi\psi}$ corresponding to higher eigenvalues). Let $P_0 = I_{d \times d}$. At OFDM symbol $m$ we use $V_{m-1}^{(d)}$ to perform the phase noise removal in the time domain as described above. The estimated symbols in the frequency domain $\hat{s}^{(m)}(k)$ are used to remove
We note that where scenario in which there is no residual carrier offset; that is, Table I (see [41]).

In the presence of residual carrier offset there is another time

\[ R_{\psi \psi}(t) = e^{j2\pi \Delta ft} R_{\psi \psi}(t) \]

and that is optional. Figure 2 depicts this phase noise

the desired signal from the received time domain signal and estimate the phase noise process at each time by

\[ \hat{\varphi}_m(t) = e^{j\varphi(m)} \sum_{k=0}^{N-1} \hat{s}(m)(k) e^{j\omega_k t} \], \( 0 \leq t \leq T_s \). \hfill (53)

\( R_{\psi \psi} \) is updated by

\[ R_{\psi \psi}^m = (1 - \alpha) R_{\psi \psi}^{m-1} + \alpha \varphi_m^* \varphi_m. \hfill (54) \]

\( V_t^{(d)} \) is updated using the PAST algorithm as described in Table 1 (see [41]).

Note that to track the subspace we do not need to compute \( R_{\psi \psi} \) and that is optional. Figure 2 depicts this phase noise compensation scheme with subspace tracking.

**B. Residual carrier offset**

In the presence of residual carrier offset there is another time varying multiplicative noise; namely, the slowly varying residual carrier. While for small values (e.g., 1 ppm) the residual carrier does not affect the decoding, it does have a detrimental effect on the common phase error estimation since the phase is no longer fixed at the pilots. Interestingly, this residual carrier can be incorporated into the KLT representation of the phase noise process by replacing \( \psi(t) \) with the random process \( \tilde{\psi}(t) \)

\[ \tilde{\psi}(t) = e^{j2\pi \Delta ft} \tilde{\psi}(t) \]

\[ = e^{j(\tilde{\psi}(t)+2\pi \Delta ft)} \hfill (55) \]

and then tracking the subspace of \( R_{\tilde{\psi} \psi} \) instead of \( R_{\psi \psi} \). We note that \( \tilde{\psi}(mT_s) \) and \( \tilde{\psi} \) are defined similarly to the scenario where there is no residual carrier offset; that is, \( \tilde{\psi}_m = \tilde{\psi}(mT_s) \) and \( \tilde{\psi} = [\tilde{\psi}_1, \ldots, \tilde{\psi}_N]^T \). Let \( c(t) = e^{j2\pi \Delta ft} \) and \( c_m = c(mT_s) \). We can represent \( R_{\tilde{\psi} \psi} \) by the following identity

\[ R_{\tilde{\psi} \psi} = \text{diag}(c) R_{\psi \psi} \text{diag}(c)^* \hfill (56) \]

where

\[ c = [c_1, \ldots, c_N]^T. \hfill (57) \]

Let \( \lambda \) be an eigenvalue of \( R_{\psi \psi} \) and \( v \) its respective eigenvector. By the definition of \( c \), \( \text{diag}(c)^* = \text{diag}(c)^{-1} \), thus

\[ R_{\tilde{\psi} \psi} \text{diag}(c) v = \text{diag}(c) R_{\psi \psi} \text{diag}(c)^* \text{diag}(c) v \]

\[ = \text{diag}(c) R_{\psi \psi} \text{diag}(c)^{-1} \text{diag}(c) v \]

\[ = \text{diag}(c) R_{\psi \psi} v \]

\[ = \lambda \text{diag}(c) v \hfill (58) \]

and we conclude that \( \text{diag}(c) v \) is an eigenvector of \( R_{\psi \psi} \) for the eigenvalue \( \lambda \). It follows that the basis elements of \( R_{\psi \psi} \); i.e., the covariance matrix of the PN process without the residual frequency offset, are multiplied by exponentials of the form \( c(t) = e^{j2\pi \Delta ft} \). This is especially appealing when we implement the adaptive tracking of the KLT basis elements, as will be demonstrated in the simulations. Note that the subspace tracking algorithm, does not require \( \Delta f \) to be known, but it is affected by it implicitly.

**C. The asymptotic behavior of the PAST algorithm**

The asymptotic behavior of the PAST and PASTd algorithms is discussed in [47]–[49] for real valued independent identically distributed (i.i.d.) Gaussian random vectors. Under the reasonable assumption that the random process \( \psi \) is \( M \)-dependent, we can sample \( \psi \) every \( M + 1 \) samples. This will slow the convergence by a linear factor of \( M + 1 \). To use the results of [47]–[49] the problem can be represented as a real problem, using standard transformation. Let,

\[ \begin{bmatrix} \text{Re}(V) & -\text{Im}(V) \\ \text{Im}(V) & \text{Re}(V) \end{bmatrix} \]

it follows that

\[ \begin{bmatrix} \text{Re}(e^{-j\varphi}) \\ \text{Im}(e^{-j\varphi}) \end{bmatrix} = V_{2\text{rep}} \begin{bmatrix} \text{Re}(\gamma) \\ \text{Im}(\gamma) \end{bmatrix} \hfill (60) \]

and we can track the subspace of the matrix \( V_{2\text{rep}} \). Thus, assuming that the vectors \( \psi \) are circularly symmetric complex normal vector, we can utilize the results of [48], [49] and have that under mild conditions, the subspace \( V_t^{(d)} \) converges to \( V_t \) with probability 1 as \( m \to \infty \).

**VI. Experimental Analysis**

In this section we present the simulated and experimental analysis of the proposed PN compensation scheme in subsection III-A. This section is divided into three parts: Section VI-A covers simulation tests of the PN compensation scheme with no tracking. Section VI-B is dedicated to the analysis of the compensation scheme with no tracking performed on the measured PN. Last, Section VI-C is dedicated to the analysis of the tracking algorithm that was proposed in Section V. This analysis is carried out on the measured PN of Section VI-B. A detailed description of the measurements that were used to produce the figure in this sections is included in [50]. The SNR was chosen such that the phase noise is the dominant noise, and limits the reception of 256 QAM. Still, a noise figure of 7 dB is quite high and even with this strong noise the phase noise is still dominant. Additionally, PN compensation complicates the design of the receiver, therefore...
it is the most cost effective when the PN is the dominant noise which typically occurs when transmitting high order constellations (i.e., high SNR) together with multi-antenna receiver.

A. Simulation tests

We now present a simulated experiment testing the performance of the algorithm on real WLAN channels. This set of simulations assumes that the actual estimate of the eigenvectors of $R_{\psi\psi}$ is given. An example of a measured channel is depicted in Figure 3. The transmitted power was 10 dBm, the assumed noise figure was 7 dB, the thermal noise was $-174$ dBm/Hz and the bandwidth was 20 MHz. The phase noise process was generated using a second order Chebychev type I and a PSD of the phase noise process is depicted in Figure 4. The standard deviation of the phase noise was $\sigma_\phi = 3^\circ$. At each time we used two receive channels and tone numbers $1-7, 21, 43, 58-64$ at each of the two receivers for pilot symbols $n_p = 16$. The OFDM had 64 tones; i.e., $N = 64$ and the modulation was 256 QAM.

Figure 5 depicts the dependence of the residual Error Vector Magnitude (EVM) on the number of basis elements using KL basis vectors and DFT basis vectors which the algorithm that was presented in [30]. The KL eigenvectors were computed based on 10 OFDM symbols. We averaged the EVM over 300 measured pairs of channels ($1 \times 2$ systems, i.e., one transmitting antenna and two receiving antennas). For each channel we averaged the EVM over 100 OFDM symbols. Note that for a large number of basis elements, the EVM may be larger due to the insufficient number of equations when
more basis elements are used than pilot symbols. The large gain of our compensation scheme compared to the DFT basis presented in [30] is clearly visible. Furthermore, the maximal tolerated EVM for 256 QAM for 5G is -32dB for a SISO transmission. The typical values for EVM are between -40dB and -35dB for MIMO channels. As can be seen, the method of [30] narrowly meets the requirements while our compensation scheme is well in the desired region.

To obtain performance under good phase noise and channel conditions we repeated the experiment with a simulated phase noise with standard deviation of $\sigma_\phi = 0.7^\circ$ and with the channel attenuation reduced by 5dB compared to Figure 3. The results are presented in Figures 6-7. Even in this case there was a substantial gain achieved by cancelling the phase noise although the phase noise performance was reasonable even with CPE compensation alone.

We also simulated the performance of our phase noise compensation scheme and that of Casas et al. [30], as a function of the of the standard deviation of the phase noise. That is, we chose a fixed number of 8 basis elements, and simulated the EVM as a function of the phase noise standard deviation. The results of this simulation are depicted in Figure 8. It is clear that our scheme outperformed the scheme presented in [30]. Specifically, using DFT basis, a phase noise with standard deviation of $3^\circ$ is the maximal phase noise which can be tolerated for a constellation of 256QAM, whereas our scheme can tolerate up to $8^\circ$ standard deviation phase noise. As before we used in this simulation 300 pairs of measured channels. For each pair channel we used 100 OFDM symbols; the phase noise process was produced for each of these channels as described at the beginning of Section VI-A.

In addition, we performed a coded simulation using a convolutional code with rate $\frac{1}{2}$, interleaver and soft-decision Viterbi decoding; its results are depicted in Figure 9. As before we used in this simulation 300 pairs of measured channels. For each pair channel we used 100 OFDM symbols; in total we used $7.68 \times 10^6$ uncoded bits. The phase noise process was produced for each of these channels as described at the beginning of Section VI-A. As can be seen from Figure 9 phase noise compensation is essential for correct decoding at the receiver. Further, the KL compensation scheme continued to outperform the DFT compensation scheme with bit error rate (BER) of less than $10^{-6}$.

The last simulation of this section depicts the performance of the phase noise compensation scheme for multi user MIMO system (see Section IV-B) with two single antenna transmitters
and two antenna receiver (2 × 2 system); Figure 10 depicts an example of the channels of a 2 × 2 communication system. The transmitted power of each transmitting antenna was 10 dBm. For each of the transmitting antennas we generated a phase noise process with standard deviation of 1° which represents the residual phase noise process at the transmitter. The two phase noise processes at the transmitters were generated independently, and were also independent of the receiver phase noise process. We compared the performance of the system with and without transmitter phase noise. The results of the simulation are presented in Figure 11, the receiver phase noise standard deviation was varied from 1° to 6°. We can clearly see from Figure 11 that when the receiver phase noise standard deviation is greater than 1°, the performance of the phase noise compensation schemes with or without transmitter phase noise are very close to one another. Clearly, the method presented in this paper does not break down in the presence of independent residual transmitter phase noise. Note that as before we used in this simulation 300 pairs of measured channels. For each pair channel we used 100 OFDM symbols; the receiver phase noise process was produced for each of these channels as described at the beginning of Section VI-A.

B. Measured phase noise analysis

In the second set of simulated experiments we used measured phase noise samples. The measured signal comprised a sine wave at 5 MHz, sampled at 40 MHz and then filtered to remove the original sine wave of 5 MHz, thus only the phase noise remained. The PSD of the phase noise is depicted in Figure 12. We repeated the experiment above with samples of the measured phase noise. The results are depicted in Figure 13. There was a clear gain of 5 dB for 5 basis elements and above.

C. Simulations of the tracking algorithm for measured phase noise

In this section we analyze the tracking capability of the PAST algorithm combined with decision direction. We used the same system as in previous sections with the measured phased noise (scaled to 3.5° total phase noise) and 4 basis elements. However, we also added a 1 ppm residual carrier to model carrier offset for a carrier frequency of 5 GHz. It is clear (see Figure 14) that our tracking scheme yields better phase and residual carrier compensation of 5 dB, compared to

3We note that the two spikes in the PSD of the measured phase noise process are due to residual of the sine and its harmony of the sine wave at 5 MHz.
the other methods, i.e., the DFT of [30] and a simple CPE compensation.

To test the stationarity of the basis elements after convergence was achieved; we repeated the experiment when no residual carrier was present and limited the training phase to 300 symbols. This is important for the quality of the initial basis, based on previous estimates. We performed two simulations whose results appear in Figure 15 and Figure 16. Figure 15 depicts the results of the simulation averaged over 300 pairs of receive channels. Figure 16 depicts the results of the simulation averaged over 300 pairs of receive channels. It is clear that the basis vectors were relatively stationary since the EVM was fixed for the duration of the next 2700 symbols. Further, the estimates were quite good and the mean EVM remained constant. This suggests that our stationarity assumption was sufficiently good. Note that both common phase removal and DFT based compensation were not as good and led to a 2 – 4dB loss.

To depict the advantages of tracking we performed the following simulation over the measured channels, we set the residual carrier to be 5 ppm and simulated phase noise of 3.5°, we averaged the results over 300 pairs of channels. We stopped the tracking of the limited tracking after 250 OFDM symbols and performed the tracking for the “Tracking using PAST” line during the whole simulation. As can be seen from Figure 17, tracking dramatically improves the behavior of the phase noise compensation scheme presented in this paper.

VII. Conclusion

We presented a novel phase noise estimation technique. This technique leads to considerable reduction in phase noise and in particular works very well for strong phase noise. The specific basis proposed in this paper (using the eigenvectors of the noise correlation process) accounts primarily for its success compared to previous work. We identified several possible extensions of the method to multi user MIMO systems as well as online calibration and exploitation of null tones. Finally, we tested the possibility of decision directed tracking of the basis vectors. The tracking results using measured phase noise and residual carrier offset suggest that our stationarity assumption also holds.
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Fig. 17: Comparison of the performance of phase noise compensation schemes with or without tracking.


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