

# Design of Precoded Waveform With Low PAPR for High Mobility Communication Systems

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**Abstract**—In this letter, the low peak to average power ratio (PAPR) precoded waveform design for high mobility communication systems is studied. Based on the time domain equivalent received signal model, a generalized transceiver framework for precoded waveforms is developed, where the computational complexity of the equalizer can be reduced by utilizing the sparse time-delay (TD) domain channel matrix. Further, a minimum signal to interference plus noise ratio (SINR) maximization precoded waveform scheme is proposed, where the precoder and equalizer are calculated jointly using an alternating optimization algorithm. The proposed precoded waveform also holds low PAPR by constraining the  $l_n$ -norm of the precoder matrix. Simulation results demonstrate that the proposed precoded waveform can achieve better performance than existing waveforms in terms of both PAPR and bit error rate (BER).

**Index Terms**—Doubly selective fading, precoded waveform design, PAPR restriction, alternating optimization.

## I. INTRODUCTION

UNDER high mobility communication scenarios, multipath and Doppler effects usually result in doubly selective fading channels. Conventional multi-carrier waveforms like orthogonal frequency division multiplexing (OFDM) are susceptible to doubly selective fading, and thus suffer from severe bit error rate (BER) performance degradation. Furthermore, the high peak to average power ratio (PAPR) is another demerit of OFDM. Existing studies have proposed schemes based on channel coding [1], [2], and precoding [3], [4], [5], [6] to reduce PAPR. Compared to coding-based approaches, precoding has been regarded as a good solution to improve the overall BER performance under selective fading without redundant information [7].

If channel state information (CSI) is available at the transmitter, CSI based precoding schemes can be implemented, which aim at compensating for the channel impairments at the transmitter. In [8] and [9], the authors proposed precoders for OFDM systems with zero forcing (ZF) and minimum mean square error (MMSE) equalizers to minimize the BER

under slow fading channel. The authors of [10] proposed a precoder towards BER minimization for OFDM systems with insufficient cyclic prefix (CP). For fast fading channel, precoding was studied in [11], where the authors proposed an eigenvalue decomposition (EVD) based adaptive precoder for an OFDM system with MMSE frequency domain block-wise equalization, which aims at maximizing the minimum signal to noise ratio (SNR) to improve the BER performance.

Nevertheless, little attention was paid to PAPR restriction in existing studies on CSI based precoded waveforms [8], [9], [10], [11]. Additionally, in the presence of fractional Doppler shifts, the Doppler spreads to all Doppler bins and results in a dense frequency-Doppler domain channel matrix, which increases the complexity of the frequency domain block-wise channel equalizer in [11]. These challenges motivate us to develop a novel transceiver model and design a novel precoding scheme to overcome the shortcomings of the existing schemes.

The contributions of this letter are summarized as:

- 1) A generalized transceiver framework for time domain precoded waveforms is developed, where existing waveforms can be generated. By utilizing the sparse TD domain channel matrix, the computational complexity of the channel equalizer can be reduced, even with fractional Doppler shifts.
- 2) Based on the generalized framework, a novel precoded waveform scheme for minimum SINR maximization is proposed. The proposed waveform holds low PAPR by constraining the  $l_1$ -norm of the precoder matrix. Simulation results demonstrate that the proposed waveform achieves better PAPR and BER performance than other existing waveforms.

Notations: The superscripts  $(\cdot)^T$ ,  $(\cdot)^H$  and  $(\cdot)^n$  represent the transpose, conjugate transpose, and  $n$ th-power operation, respectively.  $\sum(\cdot)$  is the summation operator.  $\delta(\cdot)$  denotes the Dirac delta function.  $\|\cdot\|_n$  denotes the  $l_n$ -norm. The notation  $\otimes$  denotes the Kronecker product. The normalized  $N$ -point discrete Fourier transform (DFT) and inverse DFT (IDFT) matrices are denoted by  $\mathbf{F}_N$  and  $\mathbf{F}_N^H$ , respectively.

## II. PRECODED WAVEFORM FRAMEWORK USING TIME DOMAIN EQUIVALENT SIGNAL MODEL

In this section, a generalized precoded waveform framework is developed based on the time domain signal model. Existing waveforms can be generated from this framework.

### A. Time Domain Equivalent Signal Model

Considering a doubly selective fading channel, the continuous-time domain received signal is given by [12]

$$r(t) = \int \int h(\tau, \nu) s(t - \tau) e^{j2\pi\nu(t-\tau)} d\tau d\nu + v(t), \quad (1)$$

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where  $s(t)$  is the continuous-time domain transmitted signal,  $v(t)$  is the additive white Gaussian noise,  $h(\tau, \nu)$  is the delay-Doppler channel response, given by

$$h(\tau, \nu) = \sum_{i=1}^P h_i \delta(\tau - \tau_i) \delta(\nu - \nu_i), \quad (2)$$

where  $P$  is the number of paths,  $h_i$ ,  $\tau_i$ , and  $\nu_i$  are the gain, delay, and Doppler shift of the  $i$ -th path, respectively.

Considering a block with  $N_{fft}$  subcarriers and  $N$  symbols, where the subcarrier bandwidth and symbol duration are denoted as  $\Delta f$  and  $T = \frac{1}{\Delta f}$  respectively. The normalized delay and Doppler of the  $i$ -th path are given by

$$l_i = \tau_i N_{fft} \Delta f, k_i = \nu_i NT. \quad (3)$$

The sampling rate  $f_s = N_{fft} \Delta f$  is set sufficient to approximate the path delays to the nearest sampling points in typical wide-band systems [13]. Therefore we do not need to consider non-integer  $l_i$  values. However integer  $k_i$  is not guaranteed since the duration  $NT \ll 1s$  in typical systems. The discrete-time signal after CP removal is given by

$$r(n) = \sum_{i=1}^P h_i e^{j2\pi \frac{k_i(n-l_i)}{N_{fft}N}} s([n-l_i]_{N_{fft}N}) + v(n), \quad (4)$$

where  $0 \leq n \leq N_{fft}N - 1$ . After serial-to-parallel conversion, the received signal can be denoted in vector form as

$$\mathbf{r} = \mathbf{H}\mathbf{s} + \mathbf{v}, \quad (5)$$

where  $\mathbf{v} \in \mathbb{C}^{N_{fft}N \times 1}$  is the noise vector and  $\mathbf{H} \in \mathbb{C}^{N_{fft}N \times N_{fft}N}$  is the TD domain channel matrix, given by

$$\mathbf{H} = \sum_{i=1}^P h_i \mathbf{\Pi}^{l_i} \mathbf{\Delta}^{(k_i)}, \quad (6)$$

where  $\mathbf{\Pi} = \text{circ}\{[0, 1, 0, \dots, 0]_{N_{fft} \times N}^T\}$  is a permutation matrix and

$$\mathbf{\Delta}^{(k_i)} = \text{diag} \left( e^{\frac{j2\pi k_i(0)}{N_{fft}N}}, e^{\frac{j2\pi k_i(1)}{N_{fft}N}}, \dots, e^{\frac{j2\pi k_i(N_{fft}N-1)}{N_{fft}N}} \right).$$

It can be observed from (6) that the number of nonzero elements of  $\mathbf{H}$  is determined by  $\sum_{i=1}^P \mathbf{\Pi}^{l_i}$ . In most practical mobile channels, the number of paths  $P \ll N_{fft}N$ , resulting in a sparse structure of  $\mathbf{H}$ . Note that for a short frame with fractional normalized Doppler shifts, the sparsity of TD domain channel in (6) still holds. Therefore, employing a TD domain channel equalizer can lead to a substantial reduction in computational complexity compared to using a frequency domain block-wise equalizer with a dense equivalent channel matrix. For example, the complexity of MMSE channel equalizer is reduced from  $\mathcal{O}((N_{fft}N)^3)$  to  $\mathcal{O}(P^2(N_{fft}N))$ .

### B. Generalized Precoded Waveform Framework

By utilizing the equivalent signal model in (5), a generalized framework of time domain precoded waveforms is developed, as shown in Fig.1, where S/P and P/S stand for serial-to-parallel and parallel-to-serial conversions, respectively. The parallel signal after precoding is denoted in vector form as

$$\mathbf{s} = \mathbf{U}\mathbf{x}, \quad (7)$$

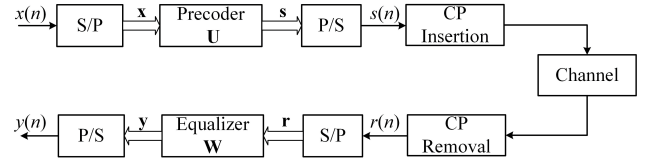


Fig. 1. Transceiver framework for precoded waveform.

where  $\mathbf{U} \in \mathbb{C}^{N_{fft}N \times MN}$  is the generalized precoder matrix, which performs precoding and upsampling to  $N$  consecutive symbols of length  $M$ . After equalization using the equalizer  $\mathbf{W} \in \mathbb{C}^{MN \times N_{fft}N}$ , the received signal is denoted as

$$\mathbf{y} = \mathbf{W}\mathbf{r} = \mathbf{W}\mathbf{H}\mathbf{U}\mathbf{x} + \mathbf{W}\mathbf{v}. \quad (8)$$

Existing waveforms like OFDM and discrete fourier transform spread OFDM (DFT-s-OFDM) can be treated as precoded waveforms and generated using such framework, where  $N = 1$  and  $\mathbf{x} \in \mathbb{C}^{M \times 1}$ , the precoder matrices are given by

$$\mathbf{U}_{\text{OFDM}} = \mathbf{F}_{N_{fft}}^H \mathbf{Z}, \quad (9)$$

$$\mathbf{U}_{\text{DFT-s-OFDM}} = \mathbf{F}_{N_{fft}}^H \mathbf{Z} \mathbf{F}_M, \quad (10)$$

respectively, where  $\mathbf{Z} \in \mathbb{C}^{N_{fft} \times M}$  is the zero padding matrix that maps  $M$  modulated symbols into  $N_{fft}$  subcarriers.

The orthogonal time frequency space (OTFS) waveform has been proposed to improve time and frequency diversity gain in fast fading channels [14]. OTFS can also be generated in the form of (7), where the transmitted and precoded symbols are vectorized as  $\mathbf{x} \in \mathbb{C}^{MN \times 1}$  and  $\mathbf{s} \in \mathbb{C}^{N_{fft}N \times 1}$  respectively. The precoder matrix  $\mathbf{U}_{\text{OTFS}} \in \mathbb{C}^{N_{fft}N \times MN}$  is derived as

$$\begin{aligned} \mathbf{U}_{\text{OTFS}} &= (\mathbf{F}_N^H \otimes \mathbf{I}_{N_{fft}})(\mathbf{I}_N \otimes \mathbf{F}_{N_{fft}}^H \mathbf{Z} \mathbf{F}_M) \\ &= \mathbf{F}_N^H \otimes \mathbf{F}_{N_{fft}}^H \mathbf{Z} \mathbf{F}_M. \end{aligned} \quad (11)$$

If the TD domain channel  $\mathbf{H}$  is perfectly known at the transmitter, CSI based precoding can be implemented. The adaptive EVD precoding [11] is reformulated based on the proposed framework as

$$\mathbf{U}_{\text{EVD}} = \mathbf{Q} \mathbf{F}_{N_{fft}}^H \mathbf{Z} \mathbf{F}_M, \quad (12)$$

where  $\mathbf{Q}$  is obtained by EVD of  $\mathbf{H}^H \mathbf{H}$ , as

$$\mathbf{H}^H \mathbf{H} = \mathbf{Q} \mathbf{\Lambda} \mathbf{Q}^H. \quad (13)$$

### III. LOW PAPR PRECODED WAVEFORM SCHEME

Although the adaptive EVD precoding scheme can achieve lower BER than non-CSI based precoding, the PAPR of the precoded waveform is as high as that of OFDM, which means it is susceptible to nonlinear distortions. This motivates us to design a novel precoding scheme that can improve the error performance while holding low PAPR.

Since the equivalent transmission matrix  $\mathbf{T} = \mathbf{W}\mathbf{H}\mathbf{U}$  that achieves minimum BER may not be strictly diagonal, there may be residual interference between received symbols. Minimal BER can be obtained by maximizing the minimum signal to interference plus noise ratio (SINR) of the received symbols. The SINR of the  $m$ -th data symbol is calculated as

$$\text{SINR}_m = \frac{|[\mathbf{W}\mathbf{H}\mathbf{U}]_{m,m}|^2}{\sigma_v^2 + \sum_{i \neq m} |[\mathbf{W}\mathbf{H}\mathbf{U}]_{m,i}|^2}, \quad (14)$$

where  $\sigma_v^2$  is the noise power. Our purpose is to design the precoder  $\mathbf{U}$  and the equalizer  $\mathbf{W}$  to maximize the minimum SINR, subject to the average power constraints and frequency domain zero index constraints of  $\mathbf{U}$ , as well as the power normalization constraint of  $\mathbf{W}$ . The problem is formulated as

$$\begin{aligned} & \underset{\mathbf{U}, \mathbf{W}}{\text{maximize}} \quad \min_{m=1, \dots, M} \text{SINR}_m \\ & \text{s.t.} \quad \|\mathbf{U}\|_2 \leq 1, \\ & \quad \|\mathbf{W}\|_2 = 1, \\ & \quad [\mathbf{F}_{N_{fft}} \mathbf{U}]_z = 0, \quad z \in \mathbf{Z}\mathbf{I}, \end{aligned} \quad (15)$$

where  $\mathbf{Z}\mathbf{I}$  contains the index of zero rows of  $\mathbf{Z}$ , since only  $M$  rows of the equivalent subcarrier mapping matrix  $\mathbf{F}_{N_{fft}} \mathbf{U}$  are set nonzero to maintain the same frequency occupancy with OFDM. Problem (15) is not convex in variables  $\mathbf{U}$  and  $\mathbf{W}$ . To this end, an alternating optimization method is proposed in this section to jointly calculate matrices  $\mathbf{U}$  and  $\mathbf{W}$ .

#### A. Precoder & Equalizer Design With the Other Given

For the precoder  $\mathbf{U}$  design with given  $\mathbf{W}$ , besides the maximization of minimum SINR, the construction of  $\mathbf{U}$  for PAPR restriction is additionally considered. From (9), (10), and (11), it can be found that the precoder matrix of DFT-s-OFDM has sparse rows when compared to OFDM and OTFS, i.e. most of the elements in each row approach 0. This means in DFT-s-OFDM, only a few input symbols in  $\mathbf{x}$  are superposed on each time domain sample of  $\mathbf{U}\mathbf{x}$ . This property motivates us to design  $\mathbf{U}$  with sparse rows. In this way, the input symbols are not constructively superposed. Only very few input symbols contribute to each output sample and the single carrier characteristics can be preserved [3], [4]. This will lead to a reduction of the peak amplitude of the output sample, which further results in a reduction of PAPR. Strict sparsity of a matrix is measured by  $l_0$ -norm. However, optimization on  $l_0$ -norm is an NP-hard problem. As an alternative, one can use  $l_1$ -norm as a convex relaxation of  $l_0$ -norm and similar sparse solutions can be obtained [15].

Typically one cannot find a  $\mathbf{U}$  that maximizes the minimum SINR while minimizing the  $l_1$ -norm as well. Alternatively, we consider maximizing the minimal SINR subject to a precoder matrix  $l_1$ -norm constraint. Therefore (15) is reformulated as:

$$\mathcal{S}(\mathcal{A}_O) = \begin{cases} \underset{\mathbf{U}}{\text{maximize}} & \min_{m=1, \dots, M} \text{SINR}_m \\ \text{s.t.} & \|\mathbf{U}\|_1 \leq \mathcal{A}_O, \\ & \|\mathbf{U}\|_2 \leq 1, \\ & [\mathbf{F}_{N_{fft}} \mathbf{U}]_z = 0, z \in \mathbf{Z}\mathbf{I}. \end{cases} \quad (16)$$

In our system, the problem (16) is not solved directly, since the optimal constraint  $\mathcal{A}_O$  is initially unknown. To this end, another strategy which minimizes the  $l_1$ -norm of  $\mathbf{U}$  given the required minimal SINR constraint  $\gamma_O$  is proposed, which is formulated as follows:

$$\mathcal{A}(\gamma_O) = \begin{cases} \underset{\mathbf{U}}{\text{minimize}} & \|\mathbf{U}\|_1 \\ \text{s.t.} & \min_{m=1, \dots, M} \text{SINR}_m \geq \gamma_O, \\ & \|\mathbf{U}\|_2 \leq 1, \\ & [\mathbf{F}_{N_{fft}} \mathbf{U}]_z = 0, z \in \mathbf{Z}\mathbf{I}. \end{cases} \quad (17)$$

According to [16], the necessary condition for (17) to be solvable is

$$\gamma_O \leq \left( \frac{\sigma_v^2}{E(|\mathbf{H}(m, n)|^2)} + \frac{M}{\text{rank}(\mathbf{W}\mathbf{H})} - 1 \right)^{-1}.$$

If  $\mathcal{A}(\gamma_O)$  is solvable, we have the following theorem:

*Theorem 1:* The  $l_1$ -norm minimization problem  $\mathcal{A}(\gamma_O)$  is the dual problem of  $\mathcal{S}(\mathcal{A}_O)$  and therefore

$$\mathcal{S}(\mathcal{A}(\gamma_O)) = \gamma_O. \quad (18)$$

In addition, the optimal solution is strictly monotonic increasing in argument  $\gamma_O$ :

$$\mathcal{A}(\gamma_O) > \mathcal{A}(\tilde{\gamma}_O), \forall \gamma_O > \tilde{\gamma}_O. \quad (19)$$

*Proof:* First, we prove (18). Assuming that  $\mathbf{U}$  and  $\mathcal{A}$  are the optimal argument and objective value of  $\mathcal{A}(\gamma_O)$ , respectively, while  $\tilde{\mathbf{U}}$  and  $\tilde{\gamma}_O \neq \gamma_O$  are the optimal argument and objective value of  $\mathcal{S}(\mathcal{A}_O)$ , respectively. If  $\tilde{\gamma}_O < \gamma_O$ , this contradicts with the optimality of  $\tilde{\mathbf{U}}$  and  $\tilde{\gamma}_O$  for  $\mathcal{S}(\mathcal{A}_O)$ , since  $\mathbf{U}$  is also feasible for it however provides a larger objective value  $\gamma_O$ . Otherwise, if  $\tilde{\gamma}_O > \gamma_O$ , then this is a contradiction for the optimality of  $\mathbf{U}$  and  $\mathcal{A}$  for  $\mathcal{A}(\gamma_O)$ , since we can always find  $0 < c < 1$  such that  $c\tilde{\mathbf{U}}$  is also feasible for  $\mathcal{A}(\gamma_O)$  however provides an objective value no larger than  $c\mathcal{A}$ .

Next, we prove (19). Assuming that  $\mathbf{U}$  and  $\mathcal{A}$  are the optimal argument and objective value of  $\mathcal{A}(\gamma_O)$ ,  $\tilde{\mathbf{U}}$  and  $\tilde{\mathcal{A}} \geq \mathcal{A}$  are the optimal argument and objective value of  $\mathcal{A}(\tilde{\gamma}_O)$  with  $\tilde{\gamma}_O < \gamma_O$ . However there always exists  $0 < c < 1$  such that  $c\mathbf{U}$  is also feasible for  $\mathcal{A}(\tilde{\gamma}_O)$  with an objective value  $c\mathcal{A} < \mathcal{A} \leq \tilde{\mathcal{A}}$ , which is a contradiction for the optimality of  $\tilde{\mathbf{U}}$  and  $\tilde{\mathcal{A}}$  for  $\mathcal{A}(\tilde{\gamma}_O)$ . ■

Using Theorem 1, we can solve  $\mathcal{S}(\mathcal{A}(\gamma_O))$  by solving  $\mathcal{A}(\gamma_O)$  for different  $\gamma_O$  values. The inversion property guarantees that the feasible  $\gamma_O$  and optimal argument  $\mathbf{U}$  for  $\mathcal{A}(\gamma_O)$  are optimal for  $\mathcal{S}(\mathcal{A}(\gamma_O))$ . It is further noticed that  $\mathcal{A}(\gamma_O)$  is unsolvable when  $\gamma_O$  is very large. Thus the maximum minimal  $\text{SINR}_m$  is found by finding the maximum feasible  $\gamma_O$  for  $\mathcal{A}(\gamma_O)$ , which can be achieved through a bisection search between a range of  $\gamma_O$ s. This procedure has been detailed in Algorithm 1. Note that the objective function of  $\mathcal{A}(\gamma_O)$  is convex and the constraints have the form of second-order cone. Thus it is formulated as a second-order cone program (SOCP), which can be solved efficiently using the CVX toolbox [17].

For the design of equalizer  $\mathbf{W}$  with given  $\mathbf{U}$ , (15) is reformulated as

$$\begin{aligned} & \underset{\mathbf{W}}{\text{maximize}} \quad \min_{m=1, \dots, M} \text{SINR}_m \\ & \text{s.t.} \quad \|\mathbf{W}\|_2 = 1. \end{aligned} \quad (20)$$

The optimal solution for (20) is achieved by using an MMSE equalizer, which is calculated as

$$\mathbf{W} = \hat{\mathbf{W}} / \|\hat{\mathbf{W}}\|_2, \quad (21)$$

where

$$\hat{\mathbf{W}} = \mathbf{U}^H \mathbf{H}^H (\mathbf{H} \mathbf{U} \mathbf{U}^H \mathbf{H}^H + \sigma_v^2)^{-1}. \quad (22)$$

<sup>1</sup>Smaller  $\epsilon_1$  leads to higher minimum SINR of the precoded waveform, however requires higher search times. The search accuracy is set as  $\epsilon_1 = 10^{-3}$  to make a tradeoff between performance and complexity.

**Algorithm 1** Precoder Design With Given Equalizer

**Input:** TD domain channel  $\mathbf{H}$ , noise power  $\sigma_v^2$ , minimum SINR search accuracy  $\epsilon_1$ , equalizer matrix  $\mathbf{W}$ .

**Output:** precoder matrix  $\mathbf{U}$ , minimum SINR  $\gamma_O$ .

- 1: Initialize the search range of SINR, the upper bound is set as  $\gamma_{\max} = 2 \left( \frac{\sigma_v^2}{E(|\mathbf{H}(m,n)|^2)} + \frac{M}{\text{rank}(\mathbf{WH})} - 1 \right)^{-1}$ , the lower bound is set as  $\gamma_{\min} = 0$  for the first iteration, otherwise set as  $\gamma_{\min} = \gamma_O^{(j-1)}$ .
- 2: Set number of search times counter  $i = 1$ .
- 3: **while**  $\gamma_{\max} - \gamma_{\min} > \epsilon_1$  **do**
- 4:   Set  $\gamma_O = (\gamma_{\max} + \gamma_{\min})/2$ .
- 5:   Solve problem  $\mathcal{A}(\gamma_O)$ .
- 6:   **if** problem  $\mathcal{A}(\gamma_O)$  is solveable, **then**
- 7:     Set  $\gamma_{\min} = \gamma_O$ .
- 8:   **else**
- 9:     Set  $\gamma_{\max} = \gamma_O$ ,  $\mathbf{U}^{(i)} = \mathbf{U}^{(i-1)}$ .
- 10:   **end if**
- 11:   Set  $i = i + 1$ .
- 12: **end while**
- 13: Set outputs  $\mathbf{U} = \mathbf{U}^{(i)}$ ,  $\gamma_O$ .

**B. Alternating Optimization Algorithm**

An alternating optimization scheme is proposed to optimize  $\mathbf{U}$  and  $\mathbf{W}$  jointly, which is detailed in Algorithm 2. We have the following proposition for the convergence of Algorithm 2:

**Algorithm 2** Joint Optimization of Precoder and Equalizer

**Input:** TD domain channel  $\mathbf{H}$ , noise power  $\sigma_v^2$ , maximum minimal SINR search accuracy  $\epsilon_2$ .

**Output:** precoder matrix  $\mathbf{U}$ , equalizer matrix  $\mathbf{W}$ .

- 1: Initialize  $\mathbf{U}^{(0)}$  that satisfies average power constraint  $\|\mathbf{U}^{(0)}\|_2 \leq 1$ , minimal SINR  $\gamma_O^{(0)} = 0$ .
- 2: Set number of iterations counter  $j = 1$ .
- 3: **while**  $\gamma_O^{(j)} - \gamma_O^{(j-1)} > \epsilon_2$  **do**
- 4:   Calculate  $\mathbf{W}^{(j)}$  using (21) and (22), where  $\mathbf{U} = \mathbf{U}^{(j-1)}$ .
- 5:   Set  $\mathbf{W} = \mathbf{W}^{(j)}$  and calculate the precoder  $\mathbf{U}^{(j)}$  and minimum SINR  $\gamma_O^{(j)}$  using Algorithm 1.
- 6:   Set  $j = j + 1$ .
- 7: **end while**
- 8: Set outputs  $\mathbf{U} = \mathbf{U}^{(j)}$ ,  $\mathbf{W} = \mathbf{W}^{(j)}$ .

*Proposition 1:* Algorithm 2 converges to a local maximum point of the minimum SINR<sub>m</sub>.

*Proof:* Let us denote the objective minimum SINR<sub>m</sub> in (15) by  $\mathcal{S}(\mathbf{U}, \mathbf{W})$ , from Algorithm 2, we have

$$\mathbf{W}^{(j)} = \arg \max_{\mathbf{W}} \mathcal{S}(\mathbf{U}^{(j-1)}, \mathbf{W}), \quad (23)$$

therefore

$$\mathcal{S}(\mathbf{U}^{(j-1)}, \mathbf{W}^{(j)}) \geq \mathcal{S}(\mathbf{U}^{(j-1)}, \mathbf{W}), \forall \|\mathbf{W}\|_2 = 1, \quad (24)$$

then we have

$$\mathcal{S}(\mathbf{U}^{(j-1)}, \mathbf{W}^{(j)}) \geq \mathcal{S}(\mathbf{U}^{(j-1)}, \mathbf{W}^{(j-1)}). \quad (25)$$

TABLE I

SIMULATION PARAMETERS FOR DIFFERENT WAVEFORMS

Parameters	Values
Carrier frequency	6 GHz
Maximum speed	1000 km/h
Maximum Doppler shift	5.54 kHz
Subcarrier spacing $\Delta f$	30kHz (for other waveforms) 120kHz (for OTFS)
Number of symbols $N$	1 (for other waveforms) 4 (for OTFS)
Symbol length $M$	48 (for other waveforms) 12 (for OTFS)
FFT length $N_{fft}$	64 (for other waveforms) 16 (for OTFS)
Modulation	QPSK
Channel fading model	TDL-A [20]
Desired delay spread $DS_{desired}$	363ns
Maximum normalized delay $l_P$	7
Maximum normalized Doppler shift $k_P$	0.2

Also, we note that

$$\mathbf{U}^{(j)} = \arg \max_{\mathbf{U}} \mathcal{S}(\mathbf{U}, \mathbf{W}^{(j)}), \quad (26)$$

and therefore

$$\mathcal{S}(\mathbf{U}^{(j)}, \mathbf{W}^{(j)}) \geq \mathcal{S}(\mathbf{U}, \mathbf{W}^{(j)}), \forall \mathbf{U} \in \mathcal{M}^{(j)}, \quad (27)$$

where  $\mathcal{U}^{(j)}$  denotes the intersection of constraints of  $\mathbf{U}$ . Note that  $\mathcal{U}^{(j-1)} \subseteq \mathcal{U}^{(j)}$  and thus

$$\mathcal{S}(\mathbf{U}^{(j)}, \mathbf{W}^{(j)}) \geq \mathcal{S}(\mathbf{U}^{(j-1)}, \mathbf{W}^{(j)}). \quad (28)$$

In conclusion of (25) and (28), we obtain

$$\mathcal{S}(\mathbf{U}^{(j)}, \mathbf{W}^{(j)}) \geq \mathcal{S}(\mathbf{U}^{(j-1)}, \mathbf{W}^{(j-1)}). \quad (29)$$

Therefore, the objective function increases at each iteration and eventually converges to a local maximum point. ■

The computational complexity of the precoder matrix is dominated by iteratively solving the SOCP problem  $\mathcal{A}(\gamma_O)$  in Algorithm 1, of which the worst-case is  $\mathcal{O}((N_{fft}M)^{3.5} \log 2(1/\epsilon_1))$  using the interior point method [18]. Further note that the channel path gains, delays, and Dopplers are almost constant during a coherence period, which is in the order of tens of milliseconds [19]. During one coherence period, the precoder matrix is updated once and applied to all symbols. With given precoder and equalizer matrices, the complexity of precoding and equalizing for each symbol block are both  $\mathcal{O}(N_{fft}M)$ .

## IV. SIMULATION RESULTS

In this section, simulation results on PAPR and BER are performed to compare the proposed precoded waveform with existing waveforms. The simulation parameters are listed in Table I. For a fair comparison, the same data length and time domain MMSE equalizer are employed for all waveforms.

The CCDF of PAPR of the precoded waveforms are shown in Fig. 2(a), where the curve for the proposed precoder is obtained by averaging 10000 different channel realizations. Under the such configurations, the average  $l_1$ -norm of the normalized precoders  $\mathbf{U}_{\text{OFDM}}$ ,  $\mathbf{U}_{\text{DFT-s-OFDM}}$ ,  $\mathbf{U}_{\text{OTFS}}$ , the adaptive EVD precoder, and the proposed precoder are 8.00, 2.94, 4.66, 7.37, and 2.51 respectively. It can be observed that



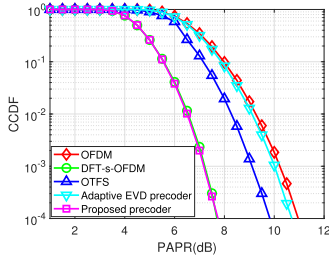


Fig. 2. CCDF of PAPR of the precoded waveforms.

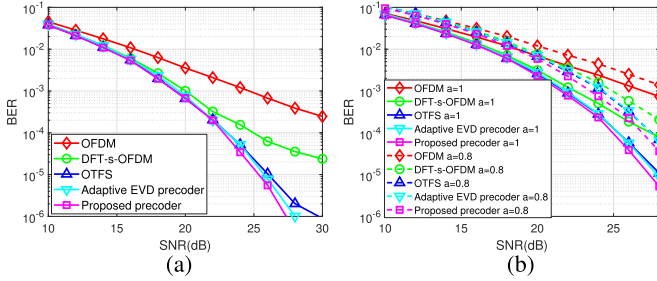


Fig. 3. BER with (a) perfect CSI (b) imperfect CSI.

the PAPR of the proposed precoder is close to DFT-s-OFDM as expected, which is about 2dB lower than that of OTFS, and about 3dB lower than those of OFDM and adaptive EVD precoder. The results validate that the PAPR is effectively reduced by using precoder matrices with lower  $l_1$ -norm. With lower PAPR, the proposed waveform is less susceptible to nonlinear distortions and has higher power efficiency.

The BER results of precoded waveforms are shown in Fig.3(a). Due to the lack of compensation for channel impairments at the transmitter, the non-CSI based precoded waveforms show higher BER than CSI based precoded waveforms. OTFS outperforms OFDM and DFT-s-OFDM since it exploits both frequency and time diversity by using the time domain MMSE equalization. It is also observed that the proposed precoder achieves lower BER than existing precoders. That is since the proposed waveform has better resistance to doubly selective fading by maximizing the minimum SINR. The PAPR and BER results indicate that the proposed method has a good potential to handle both nonlinearity and channel fading, and is thus adaptive to more severe scenarios.

To show the effectiveness of our proposed approach in practice, BER under imperfect CSI are also simulated and shown in Fig.3(b). The channel matrix is scaled by a correlation factor  $a$  and then added with a Gaussian random matrix  $\varepsilon$ . The variance of  $\varepsilon$  is  $P_\varepsilon = \frac{\sigma_\varepsilon^2}{N_p}$ , where  $N_p = 64$  is the number of pilots. It is observed that the proposed precoder keeps better performance than other existing waveforms. The results verify that the proposed precoder is still competent even with imperfect CSI.

## V. CONCLUSION

In this letter, we have developed a generalized precoded waveform framework based on the TD domain channel response, which can reduce the complexity of the equalizer. Existing waveforms can be generated from our framework.

Further, based on knowledge of the channel, a joint precoder and equalizer optimization scheme is proposed to maximize the minimum SINR while holding low PAPR. Simulation results verify that the the proposed precoded waveform achieves better PAPR and BER performance than existing waveforms.

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