Simultaneous Communication and Tracking in Arbitrary Trajectories via Beam-Space Processing

Fernando Pedraza∗, Saeid K. Dehkordi∗, Mari Kobayashi†, Giuseppe Caire∗
∗Technical University of Berlin, Germany
†Technical University of Munich, Germany
Emails: {f.pedrazanieto, s.khalilidehkordi, caire}@tu-berlin.de, mari.kobayashi@tum.de

Abstract—In this paper, we develop a beam tracking scheme for an orthogonal frequency division multiplexing (OFDM) Integrated Sensing and Communication (ISAC) system with a hybrid digital analog (HDA) architecture operating in the millimeter wave (mmWave) band. Our tracking method consists of an estimation step inspired by radar signal processing techniques, and a prediction step based on simple kinematic equations. The hybrid architecture exploits the predicted state information to focus only on the directions of interest, trading off beamforming gain, hardware complexity and multistream processing capabilities. Our extensive simulations in arbitrary trajectories show that the proposed method can outperform state of the art beam tracking methods in terms of prediction accuracy and consequently achievable communication rate, and is fully capable of dealing with highly non-linear dynamic motion patterns.

Index Terms—Integrated Sensing and Communication, Beam Tracking, OFDM, Hybrid Digital Analog.

I. INTRODUCTION

ISAC has emerged as a key enabler for 5G and beyond wireless systems to deal with the challenging requirement in terms of spectral efficiency, localization, as well as power consumption [1]. In particular, the waveform design for ISAC, sharing common hardware and spectrum, has been extensively studied in the literature (see e.g. [2] and references therein). In this context, OFDM has been widely investigated as an ISAC waveform, in virtue of its availability in wireless communication systems and the capability to achieve accurate radar estimation performance [3], [4].

Most of the envisioned ISAC applications are expected to operate in mmWave [4]. A key feature of the mmWave channel that motivates this design is its sparsity in the beamspace domain [5], which connects the scattering conditions with the geometry of the environment. Due to this property, accurate channel state information is needed in order to efficiently operate in the mmWave channel. Since acquiring this information leads to overhead, methods for dynamically inferring the channel state given past observations are currently an active area of research, often under the name of beam tracking [6]–[11]. Another key issue in mmWave are the large bandwidths involved, resulting in the need for very high sampling rates and a subsequent raise in receiver complexity and energy consumption. This problem is exacerbated when large antenna arrays are considered and the received signal is sampled and demodulated at each antenna output.

Most of the previous approaches to beam tracking [7]–[10] consider channel sounding strategies, where pilot signals are necessary to track the channel, therefore experiencing overhead. ISAC systems eliminate this problem by considering a radar receiver co-located with the transmitter, such that the transmitted waveform can carry information while being perfectly known upon reception. However, many of the current works in this framework suffer from some practical limitations. These include (i) the use of fully digital receiver architectures incurring high energy consumption [6], [11], [12], (ii) the restriction to very simple trajectories such as straight roads [6], [13], [14], and (iii) the assumption of perfect matched filtering, thus neglecting the effect of discrete sampling in time and frequency [6], [11].

In this paper, we extend our previous work on beam refinement with HDA architectures [15] to a tracking scenario by combining it with recent ideas from the literature. Our method outperforms other existing approaches by showing that well designed low dimensional observations are sufficient to achieve almost optimal tracking performance, therefore alleviating the hardware complexities in mmWave. Furthermore, we evaluate the proposed method in highly non-linear environments without the commonly used assumption of perfect matched filtering. Our simulation results demonstrate that it yields excellent performance when accurate mobility models are not available.

II. SYSTEM MODEL

We consider an ISAC scenario where a Base Station (BS) with $N_{rx}$ antennas and $N_{rf}^{rx}$ RF chains, co-located with a radar receiver equipped with $N_{rx}$ antennas and $N_{rf}^{rx}$ RF chains, tries to communicate with $K$ users, while simultaneously tracking their positions over time. We let the system operate in the mmWave band. Our model assumes that the BS has coarse knowledge of the angular locations of the $K$ users at each measurement epoch. Such knowledge can be acquired by a beam alignment method (see e.g. [16]) if a new user enters the coverage region of the BS, or predicted from previous observations for tracked users, as explained in Section III. This section focuses on a single measurement step, therefore we drop the measurement index to simplify notation.

A. Channel model

Assuming the $K$ users are well separated in space, we adopt the widely used mmWave radar channel model (see e.g. [4])
where the received echoes arrive from the line-of-sight (LoS) direction corresponding to each user, i.e.

$$\mathbf{H}(t, \tau) = \sum_{k=1}^{K} h_k \mathbf{b}(\phi_k) a^\nu(\phi_k) \delta(\tau - \tau_k)e^{j2\pi \nu_k t},$$  \hspace{1cm} (1)

where $h_k, \tau_k, \nu_k$, and $\phi_k$ are respectively the complex channel coefficient, delay, Doppler and angle of arrival (AoA) of the $k$-th user. Notice that, due to the co-location of BS and radar receiver, the angles of departure and arrival coincide. For simplicity, we focus on uniform linear arrays (ULA), such that their array response vectors have elements given by

$$[a(\phi_k)]_i = e^{i\pi(i-1) \sin(\phi_k)}, \quad i = 0, \ldots, N_{tx} - 1$$  \hspace{1cm} (2)

$$[b(\phi_k)]_i = e^{i\pi(i-1) \sin(\phi_k)}, \quad i = 0, \ldots, N_{rx} - 1$$  \hspace{1cm} (3)

According to the radar equation [17], the channel coefficient $h_k$ satisfies

$$|h_k|^2 = \frac{\lambda^2 \sigma_{\text{RCS}, k}}{(4\pi)^3 d_k}$$  \hspace{1cm} (4)

where $\lambda$ is the wavelength at the central (carrier) frequency, and $\sigma_{\text{RCS}, k}$ and $d_k$ are respectively the radar cross section (RCS) and range of the $k$-th user at a given epoch.

**B. Signal model**

We consider OFDM as our modulation scheme since it is one of the standardized waveforms for mmWave systems and due to its good applicability as an ISAC waveform [4]. In particular, we use the OFDM pulse shape

$$p_{n,m}(t) = \text{rect}\left(\frac{t - nT_0}{T_0}\right) e^{j2\pi n \Delta f(t - T_{cp} - nT_0)},$$  \hspace{1cm} (5)

where $\text{rect}(x)$ is a function taking value 1 when $0 \leq x \leq 1$ and 0 elsewhere, $\Delta f$ is the subcarrier spacing, $T_{cp}$ is the cyclic prefix duration, and $T_0 = 1/\Delta f + T_{cp}$ is the total symbol duration including cyclic prefix. Considering $N$ symbols and $M$ subcarriers, the transmitted OFDM frame is given by

$$s(t) = \frac{1}{\sqrt{K}} \sum_{k=1}^{K} \mathbf{f}(\phi_k) \sum_{n=0}^{N-1} \sum_{m=0}^{M-1} \zeta_k[n,m] p_{n,m}(t)$$  \hspace{1cm} (6)

where $\{\tilde{\phi}_k\}$ is a set of predicted AoAs, $\mathbf{f} : [-\pi/2, \pi, 2] \to \mathbb{C}^{N_{rx}}$ is a beamforming function that generates a unit norm beam pointing towards the intended user, and $\zeta_k[n,m]$ is the $n$-th information symbol intended to user $k$ sent over subcarrier $m$. For simplicity, we considered $K \leq N_{rx}^l$ in (6). As it is common in the literature [18], we will make the assumption

$$\Delta f \gg \nu_{\text{max}},$$  \hspace{1cm} (7)

where $\nu_{\text{max}}$ is the maximum Doppler frequency to be expected in the channel.

The received backscattered signal in the antenna plane and in the absence of noise is given by

$$y(t) = \frac{1}{\sqrt{K}} \sum_{k=1}^{K} h_k \mathbf{b}(\phi_k) a^\nu(\phi_k) s(t - \tau_k)e^{j2\pi \nu_k t}$$  \hspace{1cm} (8)

$$\approx \frac{1}{\sqrt{K}} \sum_{k=1}^{K} h_k \mathbf{b}(\phi_k) a^\nu(\phi_k) \mathbf{f}(\hat{\phi}_k')$$  \hspace{1cm} (9)

$$\approx \frac{1}{\sqrt{K}} \sum_{k=1}^{K} h_k \mathbf{b}(\phi_k) a^\nu(\phi_k) \mathbf{f}(\hat{\phi}_k)$$  \hspace{1cm} (10)

where the last step follows from the approximation $|a^\nu(\phi_k)\mathbf{f}(\hat{\phi}_k')| \approx 0$ for $k' \neq k$, which is accurate in massive MIMO systems when the users are spatially separated and the predictions are close to the true value [19].

Aiming to reduce hardware complexity and energy consumption at the radar receiver, we process the received signal $y(t)$ by a reduction matrix before sampling. In order to be able to produce super-resolution angle estimates, we let the radar receiver focus on a single user $k^*$ for each OFDM frame, and estimate different users sequentially in time. We achieve this by tuning the reduction matrix $U_{k^*} \in \mathbb{C}^{N_{rx} \times N_{rx}^l}$ as will be described later. Then, after standard OFDM processing (see e.g. [3]) and including noise, the sampled signal when focusing on user $k^*$ is given by

$$y_{k^*}[n,m] = U_{k^*}^n \left( \frac{1}{\sqrt{K}} \sum_{k=1}^{K} h_k e^{j2\pi nT_0 \nu_k - nT_0} b(\phi_k) a^\nu(\phi_k) \mathbf{f}(\hat{\phi}_k') \zeta_k[n,m] + \mathbf{w}[n,m] \right)$$  \hspace{1cm} (11)

$$= U_{k^*}^n \left( \sum_{k=1}^{K} h_k g_{k,k^*} b(\phi_k) \tilde{\zeta}_k[n,m] + \mathbf{w}[n,m] \right)$$  \hspace{1cm} (12)

$$\approx U_{k^*}^n h_{k^*} g_{k,k^*} b(\phi_{k^*}) \tilde{\zeta}_{k^*}[n,m] + \mathbf{w}[n,m],$$  \hspace{1cm} (13)

where in (12) we defined $g_{k,k^*} = \frac{1}{\sqrt{K}} a^\nu(\phi_k) \mathbf{f}(\hat{\phi}_k')$ and $\tilde{\zeta}_{k^*}[n,m] = \zeta_k[n,m] e^{j2\pi (nT_0 \nu_k - mT_{cp})}$, $\mathbf{w}[n,m] \in \mathbb{C}^{N_{rx}}$ is white Gaussian noise with variance $\sigma_n^2$, and the approximation in (13) follows from designing $U_{k^*}$ such that $\|U_{k^*}^n h_{k^*} b(\phi_{k^*})\| \approx 0$ for $k \neq k^*$.

**III. PROPOSED TRACKING SCHEME**

In this section, we describe the two main components of our tracking scheme. First, we estimate parameters of interest from the sampled signal $y_{k^*}[n,m]$. For our tracking scheme, we are interested in the ranges $\{d_{l,k}\}$ and angles of arrival $\{\phi_{l,k}\}$ of the different users, where $l$ indexes measurements sequentially.
Algorithm 1 Beam Tracking

Input: Refresh period $\Delta T$, predicted angles $\{\tilde{\phi}_{0,k}\}$ at $t = 0$.

\[ l \leftarrow 0 \]

for $k^* = 1, \ldots, K$ do [1]

Transmit beamformed OFDM frames towards all directions $\{\tilde{\phi}_{l,k}\}$ in the predicted set as shown in Section II

Obtain estimate $\{\hat{d}_{l,k^*}, \tilde{\phi}_{l,k^*}\}$ as shown in Section III-A

Predict $\hat{\phi}_{l+1,k^*}$ as shown in Section III-B

end for

\[ l \leftarrow l + 1 \]

Radar receiver idle until $t = l\Delta T$

end loop

in time. However, due to the nature of our estimator, we also produce estimates of the Doppler frequencies $\{\nu_{l,k}\}$, which could be used for other purposes outside of the scope of this paper. Then, we update the set of predicted angles $\{\tilde{\phi}_{l,k}\}$ based on the history of estimations. The overall procedure is summarized in Algorithm 1.

A. Radar Parameter Estimation

For this section, we will focus on estimating the parameters of user $k^*$ at the $l$-th measurement round and will drop the indices to simplify notation. The measurement model is extensively described in [15], therefore only a brief summary will be provided here.

We define the reduction matrix $U$ as

\[ U = \text{D}(\tilde{\phi})\Psi, \]

where $\text{D}(\tilde{\phi}) \triangleq \text{diag}(1, e^{j\pi\sin(\tilde{\phi})}, \ldots, e^{j\pi(N_{rx}-1)\sin(\tilde{\phi})})$ is a tunable network of phase shifters and $\Psi$ is a fixed network of beamformers pointing towards the broadside direction (i.e. $\phi = 0$), satisfying $U^*U = \frac{1}{N_{tx}N_{rx}}I_{N_{tx}N_{rx}}$, where $I_N$ is the identity matrix of rank $N$. In particular, we let the columns of $\Psi$ be obtained as the first $N_{tx}^2$ Slepian sequences of length $N_{rx}$ and any user defined time-bandwidth product. A brief description of Slepian sequences can be found in [15] and more extensive treatment in [20].

Using this design, we can obtain the sample covariance matrix from $N \times M$ samples

\[ \hat{R} = \frac{1}{NM} \sum_{n=0}^{N-1} \sum_{m=0}^{M-1} y[n, m]y^*[n, m] \]

\[ = P_{tx}|h|^2|g_{tx}|^2\Psi^*\Psi\mathbf{b}(\phi')\mathbf{b}(\phi')\Psi + \sigma_n^2 I_{N_{tx}N_{rx}}, \]

where $P_{tx} \triangleq \mathbb{E} [\|\xi[n,m]\|^2]$ is the average power of the transmitted signal and $\phi' \triangleq \sin^{-1}(\sin(\phi) - \sin(\tilde{\phi}))$. Given the structure of $\hat{R}$, we can process it via spectral methods such as MUSIC [20] in order to obtain $\tilde{\phi}$ as a super-resolution estimate of $\phi$.

Finally, we can use the signal model in (13) together with our estimate of $\phi$ to obtain a least squares estimate of $\tau$ and $\nu$. To do so, we solve

\[ (\hat{\tau}, \hat{\nu}) = \arg\min_{(\tau, \nu)} \sum_{n, m} \left| y[n, m] - h_{tx}U^*b(\tilde{\phi})\xi[n, m] \right|^2, \]

which, as shown in [15], is equivalent to solving

\[ (\hat{\tau}, \hat{\nu}) = \arg\max_{(\tau, \nu)} \sum_{n, m} y'[n, m]e^{-j2\pi(nT_{\nu}-m\Delta f\tau)}, \]

where $y'[n, m] \triangleq b^*\mathbf{U}_{y[n,m]}$. Problem (17) can be efficiently solved by applying FFT and finding a peak in a 2D grid. We can estimate the range $d$ as

\[ \hat{d} = \frac{c\hat{\tau}}{2}, \]

where $c$ is the speed of light.

B. Next Angle Prediction

In this section, we summarize the key points of the beam prediction step in [11] and propose a combination of their simple tracking equations with the radar receiver introduced in Section III-A that yields an overall improved performance. By focusing still on a single user, we will introduce the time index $l$ to capture the prediction based on the past observations.

The approach in [11] aims to predict the coordinates of the user of interest given the kinematic state in the last three measurement epochs. In particular, focusing on the $x$ coordinate,

\[ x_{l+1} - x_{l} = v_x,1\Delta T + a_x,1\Delta T^2/2, \]

\[ x_{l} - x_{l-1} = v_x,l-1\Delta T + a_x,l-1\Delta T^2/2, \]

\[ x_{l-1} - x_{l-2} = v_x,l-2\Delta T + a_x,l-2\Delta T^2/2, \]

\[ v_x,l - v_x,l-1 = a_x,l-1\Delta T, \]

\[ v_x,l-1 - v_x,l-2 = a_x,l-2\Delta T \]

where $x_{l}$, $v_x,l$ and $a_x,l$ are respectively the $x$-th coordinate, the velocity and the acceleration in the $x$-th coordinate at the $l$-th measurement step, and $\Delta T$ is the interval between measurements corresponding to the same user. In this work, we choose $\Delta T \gg T_0$ for two reasons. First, since different users are estimated sequentially in time, the number of users this system can accommodate is upper bounded by $[\Delta T/T_0]$, where $[x]$ indicates the largest integer not greater than $x$. Second, $T_0$ can be too short for any non-negligible motion to happen. Therefore, using a much longer $\Delta T$ does not significantly affect performance, while greatly simplifying hardware operation by reducing the rate at which beams must be reconfigured.

By setting the assumption that the acceleration does not significantly change in three measurement steps (i.e. $a_x,l-2 \approx a_x,l-1 \approx a_x,l$), the system of equations in (19) can be solved for $x_{l+1}$ yielding

\[ x_{l+1} = 3x_{l} - 3x_{l-1} + x_{l-2} \approx 3\hat{x}_{l} - 3\hat{x}_{l-1} + \hat{x}_{l-2}. \]
where, keeping our notation consistent, \( \hat{x} \) represents an estimated value (e.g. by following the processing in Section III-A\(^2\)) and \( \hat{x} \) represents a predicted value.

The derivation holds verbatim for the \( y \)-th coordinate, so we can obtain a prediction for the next AoA from the relation between the \( x \) and \( y \) coordinates as

\[
\hat{x}_{l+1} = \tan^{-1} \left( \frac{3\hat{x}_l - 3\hat{x}_{l-1} + \hat{x}_{l-2}}{3\hat{y}_l - 3\hat{y}_{l-1} + \hat{y}_{l-2}} \right).
\]  

(21)

### IV. Numerical Results

We compare our approach with the one proposed in [11] that assumes a fully digital receiver (i.e. \( N_{\text{rf}} = N_{\text{tx}} \)) and considers estimation via local linear approximation [11] Section III.B. We remark here that our method, by operating in the beam-space domain, overcomes two major issues associated with fully digital architectures. First, it greatly reduces hardware complexity since the wideband signal is sampled by only \( N_{\text{rf}} \) A/D converters, where we can in general have \( N_{\text{rf}} \ll N_{\text{tx}} \) as will be shown in this section. Second, it should be noted that the raw samples at the antenna outputs are extremely noisy due to the large pathloss of the two-way mmWave channel. Our method projects the high dimensional signal into a suitable low dimensional subspace where most of the information content is preserved but noise from uninteresting directions is rejected, thus resulting in a much higher SNR per sample. Slepian sequences provide a systematic way to define such a subspace by procuring an orthonormal basis upon which signals coming from directions of interest have maximal projection, whereas those coming from any other angular region are approximately orthogonal. We also note that we had to assume perfect matched filtering for the method in [11] to perform adequately, whereas our method is robust to small processing errors (e.g. those caused by discrete sampling in time and frequency). A detailed description of what this assumption entails can be found in Appendix A. For the sake of completeness, we also include performance evaluation for the method in [11] when no perfect matched filtering is assumed. Furthermore, for ease of evaluation, in the presented results we focus on a single user scenario (i.e. \( K = 1 \)). Table I summarizes the parameters used in our simulations.

| Parameter     | Value                |
|---------------|----------------------|
| \( N \)       | 64                   |
| \( M \)       | 512                  |
| \( N_{\text{tx}} \) | 64                   |
| \( N_{\text{rx}} \) | 64                   |
| \( \Delta f \) | 1 MHz                |
| \( \Delta T \) | 100 ms               |
| \( f_c \)     | 60 GHz               |
| \( d_{\text{max}} \) | 100 m               |
| \( P_{\text{tx}} \) | 50 mW               |
| Noise PSD     | \( 2 \times 10^{-21} \text{ W/Hz} \) |
| \( v_{\text{max}} \) | 30 m/s              |

![Fig. 1. Main road scenario and trajectory considered.](image1)

![Fig. 2. Achievable spectral efficiency as a function of the time step for the selected trajectory.](image2)

Noise power spectral density

\[
\delta_N = \text{log}_2 \left( 1 + \left( \frac{\lambda}{4\pi d_i} \right)^2 \frac{P_{\text{tx}}|g_{\text{tx},i}|^2}{N_0(M\Delta f)} \right),
\]  

(22)

where the approximation follows from using an empirical estimation of the expectation and \( N_0 \) is the noise power spectral density at the user. The results are presented in Fig.2 and show how our method is able to quickly recover from the non-smooth changes of direction while the approach in [11] loses track of the user and is not able to find it again.

In order to average results over different mobility patterns, we generate trajectories where a user moves within the road structure defined in Fig.1 but now is able to take any of the paths, and where the speed at each time step is sampled from a random process. Since now there is no association between time step and kinematic state, we illustrate performance as a function of the position by means of a heat map. This is shown in

\(^2\)Notice that estimates of \( x \) and \( y \) can be directly obtained from estimates of \( d \) and \( \phi \) using basic trigonometry.
point of each path for different antenna front end configurations. This is presented in Fig. 4 where it can be seen that as few as 4 RF chains are enough to achieve close to optimal performance. This result suggests that projecting the high dimensional signal at the antenna plane onto a much lower dimensional space preserves most of the information necessary for estimation, provided that the considered subspace is adequately selected. The figure also shows how the gap between the achieved results and the upper bound becomes larger when the number of antennas increases. This can be explained by the fact that larger arrays generate narrower beams, resulting on a higher sensitivity to small pointing errors.

V. CONCLUSIONS

In this paper, we have shown an efficient method to perform beam tracking in OFDM ISAC systems, based on the combination of an advanced HDA receiver and a simple tracking equation. Our close to optimal results indicate that large savings in computation and hardware complexity can be obtained without sacrificing performance. Moreover, the validation of our results in complicated road geometries suggests that our method is general enough to perform well in realistic mobility scenarios.

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where \(E_t\) vary in magnitude. In practice, most systems sample a grid of\(\frac{g}{g}\) in mobility settings where delay and Doppler might be time varying. The exact matched filter output would be given by
\[
r = h_0 g(\phi_0) b(\phi_0) \int_{-\infty}^{\infty} s(t' - \tau_0) s^\ast(t' - \hat{\tau}) e^{j2\pi(\nu_0 - \nu')t' dt'},
\]
which shows that the term \(\mathcal{E}_s\) should be replaced with an expression depending on \(\Delta\tau \Delta\tau = \tau_0 - \hat{\tau}\) and \(\Delta\nu \Delta\nu = \nu_0 - \hat{\nu}\). Notice also that \(\Delta\tau\) and \(\Delta\nu\) can vary quickly in time since \(\tau_0\) \(\nu_0\) changes smoothly and \(\hat{\tau}\) \(\hat{\nu}\) changes in discrete steps. When using model (29) in lieu of (28), many of the works in the literature become unusable. In particular, works that consider that the sampled signal is described by a statistical distribution that is a perfectly known function of the AoA and the channel coefficient fail to consider this difficult to model phenomenon. Also, methods that take derivatives of the model with respect to the parameters in order to linearize the functional dependence, like the baseline used in this paper, have typically considered the formulation in (28), resulting in simple expressions that are however invalid once the perfect matched filtering assumption is lifted.

**APPENDIX A**

**THE PERFECT MATCHED FILTERING ASSUMPTION**

In this appendix, we illustrate the limitation of the assumption on perfect matched filtering frequently used in recent works. To that aim, let us focus for simplicity in a single path channel model, such that the time-varying channel matrix is given by
\[
H(t, \tau) = h_0 b(\phi_0) a^\ast(\phi_0) \delta(t - \tau_0) e^{j2\pi\nu_0 t},
\]
where \(h_0\), \(\phi_0\), \(\tau_0\) and \(\nu_0\) represent respectively the channel coefficient, AoA, delay and Doppler associated with a single user. Let us consider a generic transmitted signal \(s(t) = f s(t)\) time limited to the interval \(t \in [0, T]\), where \(f\) is a beamforming vector, such that the received signal in the absence of noise is
\[
r(t) = h_0 g(\phi_0) b(\phi_0) s(t - \tau_0) e^{j2\pi\nu_0 t},
\]
where \(g(\phi_0) = a^\ast(\phi_0) f\). The (ideal) matched filtering process would then consist in processing the received signal as
\[
r = \int_{-\infty}^{\infty} r(t') s^\ast(t' - (t - T)) e^{-j2\pi\nu t'} dt' \bigg|_{t=T+\tau_0} \bigg|_{\nu=\nu_0}
\]
\[
= h_0 g(\phi_0) b(\phi_0) \int_{-\infty}^{\infty} s(t' - \tau_0) s^\ast(t' - (t - T)) e^{-j2\pi(\nu_0 - \nu)t'} dt' \bigg|_{t=T+\tau_0} \bigg|_{\nu=\nu_0}
\]
\[
= h_0 g(\phi_0) b(\phi_0) \int_{-\infty}^{\infty} s(t' - \tau_0) s^\ast(t' - \tau_0) dt'
\]
\[
= h_0 \mathcal{E}_s g(\phi_0) b(\phi_0),
\]
where \(\mathcal{E}_s = \int_{-\infty}^{\infty} |s(t)|^2 dt\) is the energy of signal \(s(t)\). Recent works consider a noisy version of (28) as a model for the sampled signal. However, as illustrated here, explicit knowledge of the delay and Doppler of the channel was required to arrive to this expression. This assumption becomes unrealistic in mobility settings where delay and Doppler might be time varying magnitudes. In practice, most systems sample a grid of \((\tau, \nu)\) tuples and find the pair maximizes the norm of the output of the matched filter. Let us refer to such a pair as \((\hat{\tau}, \hat{\nu})\), where \((\hat{\tau}, \hat{\nu}) \neq (\tau_0, \nu_0)\) almost surely due to the finiteness of the grid.