

Estimation of Channel Parameters in UWB Impulse Radio Communication

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1 Signal Model and problem formulation

In Impulse-Based Ultra-Wideband (IR-UWB) communication, signaling is commonly realized by transmitting very short duration pulses on the order of sub-nanoseconds [1]. The transmitted signal consists of a train of such pulses at baseband, where each transmitted pulse is generally a Gaussian pulse derivative. The received signal at the receiver in the frequency domain is given by [2],

$$y(\omega) = s(\omega) \sum_{l=1}^L \alpha_l e^{-j\omega\tau_l} + b(\omega) \quad (1.1)$$

where $y(\omega)$ is the received signal, $s(\omega)$ is the transmitted pulse, α_l is the gain and τ_l is the delay of the l^{th} path. $b(\omega)$ is the zero-mean Additive White Gaussian noise.

The discretized received signal can be written as:

$$y_n \stackrel{\text{def}}{=} y(w_n) = s_n \sum_{l=1}^L \alpha_l e^{jw_n\tau_l} + b_n, \quad k = 0, \dots, N-1 \quad (1.2)$$

where $s_n \stackrel{\text{def}}{=} s(w_n)$ and $b_n \stackrel{\text{def}}{=} b(w_n)$ with $w_n = w_0 k$ and where $w_0 = \frac{2\pi}{N}$. Let $\mathbf{y} \stackrel{\text{def}}{=} (y_0, \dots, y_{N-1})$ and $\mathbf{b} \stackrel{\text{def}}{=} (b_0, \dots, b_{N-1})^T \sim \mathcal{N}(0, \sigma^2 \mathbf{I})$ be two vectors containing the N samples of y_n and b_n , respectively. From (1.1), we have

$$\mathbf{y} = \mathbf{S} \mathbf{A}_\tau \boldsymbol{\alpha} + \mathbf{b} \quad (1.3)$$

where $\boldsymbol{\alpha} \stackrel{\text{def}}{=} (\alpha_1, \dots, \alpha_{L-1})^T$ is a vector formed by the L channel gains, \mathbf{S} is a diagonal complex matrix of dimension $(N \times N)$, whose elements are the DFT of the transmitted signal, \mathbf{A}_τ is a matrix of dimension $(N \times L)$ defined by the delay-signature vectors (i.e., harmonic components) associated to each arriving delayed signal (path),

$$\mathbf{A}_\tau = (\mathbf{a}_0, \dots, \mathbf{a}_{L-1})$$

where $\mathbf{a}_k \stackrel{\text{def}}{=} \mathbf{a}_{\tau_k} = (1, e^{-jw_0\tau_k}, \dots, e^{-jw_0(N-1)\tau_k})^T$.

With the above problem formulation, we assume that the signal s is deterministic and is the 2^{nd} derivative of Gaussian pulse given by [3],

$$s(t) = E_0 \left[1 - 4\pi \left(\frac{t-t_0}{\Delta t} \right)^2 \right] \exp \left\{ -2\pi \left(\frac{t-t_0}{\Delta t} \right)^2 \right\} \quad (1.4)$$

where E_0 is the peak amplitude at the time $t = t_0$, and Δt is a nominal duration which can be chosen to adjust the pulse width.

Furthermore, the channel is assumed to be time varying Rayleigh channel (i.e., the gains are zero-mean complex circular Gaussian but with the delays constant during the observation period) and the number of paths (i.e., L) is known.

Now we note that $\boldsymbol{\theta} \stackrel{\text{def}}{=} (\boldsymbol{\tau}, \boldsymbol{\alpha})$ is to be estimated from \mathbf{y} where $\boldsymbol{\tau} \stackrel{\text{def}}{=} (\tau_1, \dots, \tau_L)^T$.

Since the path gains of the channel are assumed stochastic with covariance matrix \mathbf{R}_α of Rank equal to L , the principle of the classical Multiple Signal Classification (MUSIC) algorithm can be applied because the Rank of the subspace signal is $L = \text{Rank}(\mathbf{R}_\alpha) = \text{Rank}(\mathbf{S}\mathbf{A}_\alpha\mathbf{R}_\alpha\mathbf{A}_\alpha^H\mathbf{S})$, where we have used the fact that the matrix \mathbf{A}_α has full column rank.

Therefore, the auto-correlation of the received signal \mathbf{R}_y is estimated at the receiver from the Q snapshots of the received signal. The noise sub-space is formed by the $N - L$ eigenvectors corresponding to the $N - L$ smaller eigenvalues of auto-correlation matrix \mathbf{R}_y . The L peaks of the MUSIC spectrum correspond to the respective delays. The MUSIC spectrum is given by,

$$F(\tau) = \frac{1}{\hat{\mathbf{y}}(\tau)^H \mathbf{U}_n \mathbf{U}_n^H \hat{\mathbf{y}}(\tau)} \quad (1.5)$$

where \mathbf{U}_n is the $N \times (N - L)$ matrix associated with the $N - L$ smaller eigenvalues.

The path gains α are estimated from the MAP estimation algorithm, which is expressed as:

$$\hat{\alpha} = \arg \max_{\alpha} p(\alpha/\mathbf{y}) \quad (1.6)$$

where $p(\alpha/\mathbf{y})$ is the a posteriori probability of α . The above equation 1.6 can be equivalently expressed as:

$$\hat{\alpha} = \arg \max_{\alpha} p(\alpha/\mathbf{y}) p(\alpha) \quad (1.7)$$

where $p(\alpha)$ is the a priori density function of α .

Monte Carlo simulations need to be done to obtain the mean-square error (MSE) on the path gain and delay parameters.

2 Objective

The main objective of this project is to estimate the UWB Channel parameters, namely path gains and delays, in a Rayleigh fading channel. The deterministic path delays are first estimated using MUSIC algorithm and then the path gains are estimated by applying the Maximum A posteriori Probability estimation (MAP) technique.

References

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