NONCOHERENT LEAST SQUARES ESTIMATORS OF CARRIER PHASE AND AMPLITUDE

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ABSTRACT

We consider least squares estimators of carrier phase and amplitude from a noisy communications signal. We focus on signaling constellations that have symbols evenly distributed on the complex unit circle, i.e., M-ary phase shift keying. We show, under reasonably mild conditions on the distribution of the noise, that the least squares estimator of carrier phase is strongly consistent and asymptotically normally distributed. However, the amplitude estimator is not consistent, but converges to a positive real number that is a function of the true carrier amplitude, the noise distribution and the size of the constellation. The results of Monte Carlo simulations are provided and these corroborate the theoretical results.

Index Terms— Noncoherent detection, phase shift keying, asymptotic statistics

1. INTRODUCTION

In passband communication systems the transmitted signal typically undergoes time offset (delay), phase shift and attenuation. These effects must be compensated for at the receiver. In this paper we assume that the time offset has been previously handled, and we focus on estimating the phase shift and attenuation. We consider signalling constellations that have symbols evenly distributed on the complex unit circle such as binary phase shift keying (BPSK), quaternary phase shift keying (QPSK) and *M*-ary phase shift keying (*M*-PSK). In this case, the transmitted symbols take the form,

$$s_i = e^{ju_i},$$

where $j = \sqrt{-1}$ and u_i is from the set $\{0, \frac{2\pi}{M}, \dots, \frac{2\pi(M-1)}{M}\}$.

We assume that time offset estimation and matched filtering have been performed and that L noisy M-PSK symbols are observed by the receiver. The received signal is,

$$y_i = a_0 s_i + w_i, \qquad i = 1, \dots, L,$$
 (1)

where w_i is noise and $a_0 = \rho_0 e^{j\theta_0}$ is a complex number representing both carrier phase θ_0 and amplitude ρ_0 (by definition ρ_0 is a positive real number). Our aim is to estimate a_0

from y_1, \ldots, y_L . If the transmitted symbols s_1, \ldots, s_L are known a priori at the receiver then the least squares estimator is typically used,

$$\hat{a}_{uc} = \arg\min_{a \in \mathbb{C}} \sum_{i=1}^{L} |y_i - as_i|^2 = \frac{1}{L} \sum_{i=1}^{L} y_i s_i^*, \qquad (2)$$

where \mathbb{C} is the set of complex numbers and x^* and |x| denote the conjugate and magnitude of the complex number x. Viterbi and Viterbi [1] call this the *unmodulated carrier* estimator. This estimator can be used if the transmitter includes *pilot* symbols, known to the receiver, i.e. *coherent detection*.

In the paper we are interested in *noncoherent detection*, where s_1, \ldots, s_L are not known at the receiver, and must also be estimated. This estimation problem has undergone extensive prior study and is often called *multiple symbol differential detection* [1–9]. A practical approach is the least squares estimator,

$$\hat{a} = \arg\min_{a \in \mathbb{C}} \min_{s_1, \dots, s_L \in \mathcal{C}} \sum_{i=1}^L |y_i - as_i|^2,$$
(3)

where C is the set of symbols from the *M*-PSK constellation. The least squares estimator is also the maximum likelihood estimator under the assumption that the noise sequence $\{w_i, i \in \mathbb{Z}\}$ is white and Gaussian. As we show, the estimator can work well even when the noise is not Gaussian. Mackenthun [6] described an algorithm to compute the least squares estimator \hat{a} that requires only $O(L \log L)$ arithmetic operations. Sweldens [7] rediscovered Mackenthun's algorithm in 2001.

In the literature it has been common to assume that the symbols s_1, \ldots, s_L are of primary interest and the complex amplitude a_0 is a nuisance parameter. The metric of performance is correspondingly the symbol error rate, or bit error rate. While estimating the symbols (or more precisely the transmitted bits) is ultimately the goal, we take the opposite point of view here. Our aim is to estimate a_0 , and we treat the unknown symbols as nuisance parameters. This is motivated by the fact that in many modern communication systems the data symbols are *coded*. For this reason raw symbol error rate

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is not of interest at this stage. Instead, we desire an accurate estimator \hat{a} of a_0 , so that the compensated received symbols $\hat{a}^{-1}y_i$ can be accurately modelled using an additive noise channel. The additive noise channel is a common assumption for subsequent receiver operations, such as decoding. The estimator \hat{a} is also used in the computation of decoder metrics for modern decoders, and for interference cancellation in multiuser systems. Consequently, our metric of performance will not be symbol or bit error rate, but $|\hat{a} - a_0|^2$. It will be informative to consider the carrier phase and amplitude estimators separately, that is, if $\hat{a} = \hat{\rho}e^{j\hat{\theta}}$ where $\hat{\rho}$ is a positive real number, then we consider $\langle \hat{\theta} - \theta_0 \rangle^2$ and $(\hat{\rho} - \rho_0)^2$. The function $\langle \cdot \rangle$ denotes its argument taken 'modulo $\frac{2\pi}{M}$ ' into the interval $[-\pi/M, \pi/M)$, that is

$$\langle x \rangle = x - \frac{2\pi}{M} \operatorname{round} \left(\frac{M}{2\pi} x \right)$$

where round(·) takes its argument to the nearest integer. The direction of rounding for half-integers is not important so long as it is consistent. We have chosen to round up half-integers here. It will become apparent why $\langle \hat{\theta} - \theta_0 \rangle^2$ rather than $(\hat{\theta} - \theta_0)^2$ is the appropriate measure of error for the phase parameter.

The paper is organised in the following way. Section 2 describes properties of complex random variables that we need. Section 3 describes the statistical properties of the least squares estimators of carrier phase $\hat{\theta}$ and amplitude $\hat{\rho}$. We show, under some assumptions about the distribution of the noise w_1, \ldots, w_L , that $\langle \hat{\theta} - \theta_0 \rangle$ converges almost surely to zero and that $\sqrt{L}\langle \hat{\theta} - \theta_0 \rangle$ is asymptotically normally distributed as $L \to \infty$. However, $\hat{\rho}$ is not a consistent estimator of the amplitude ρ_0 . The asymptotic bias of $\hat{\rho}$ is small when the signal to noise ratio (SNR) is large, but the asymptotic bias is significant when the SNR is small. Section 5 presents the results of Monte-Carlo simulations. These simulations agree with the derived asymptotic properties.

2. CIRCULARLY SYMMETRIC COMPLEX RANDOM VARIABLES

Before describing the statistical properties of the least squares estimator, we first require some properties of complex valued random variables. A complex random variable W is said to be *circularly symmetric* if its phase $\angle W$ is independent of its magnitude |W| and if the distribution of $\angle W$ is uniform on $[0, 2\pi)$. That is, if $Z \ge 0$ and $\Theta \in [0, 2\pi)$ are real random variables such that $Ze^{j\Theta} = X$, then Θ is uniformly distributed on $[0, 2\pi)$ and is independent of Z. If the probability density function (pdf) of Z is $f_Z(z)$, then the joint pdf of Θ and Z is $f_{Z,\Theta}(z,\theta) = \frac{1}{2\pi}f_Z(z)$. If X is circularly symmetric, then for any real number ϕ , the distribution of X is the same as that of $e^{j\phi}X$.

3. STATISTICAL PROPERTIES OF THE LEAST SQUARES ESTIMATOR

The next two theorems describe the asymptotic properties of the least squares estimator. We omit the proofs due to space constraints. Proofs will be provided in a forthcomming paper.

Theorem 1. (Almost sure convergence) Let $\{w_i\}$ be a sequence of independent and identically distributed, circularly symmetric complex random variables with w_1 having finite variance and continuous pdf. Let y_1, \ldots, y_L be given by (1) and let $\hat{a} = \hat{\rho}e^{j\hat{\theta}}$ be the least squares estimator of $a_0 = \rho_0 e^{j\theta_0}$ given by (3). Let $R_i \ge 0$ and $\Phi_i \in [0, 2\pi)$ be real random variables satisfying

$$R_i e^{j\Phi_i} = 1 + \frac{w_i}{a_0 s_i},$$

and define the continuous function

$$G(x) = \mathbb{E}R_1 \cos\langle x + \Phi_1 \rangle.$$

If G(x) is uniquely maximised at x = 0 over the interval $\left[-\frac{\pi}{M}, \frac{\pi}{M}\right]$, then:

- 1. $\langle \hat{\theta} \theta_0 \rangle \rightarrow 0$ almost surely as $L \rightarrow \infty$,
- 2. $\hat{\rho} \to \rho_0 G(0)$ almost surely as $L \to \infty$.

Theorem 2. (Asymptotic normality) Under the same conditions as Theorem 1, let $f(r, \phi)$ be the joint pdf of R_1 and Φ_1 , and let

$$g(\phi) = \int_0^\infty rf(r,\phi)dr.$$

Put $\hat{\lambda}_L = \langle \hat{\theta} - \theta_0 \rangle$ and $\hat{m}_L = \hat{\rho} - \rho_0 G(0)$. Then the distribution of $(\sqrt{L}\hat{\lambda}_L, \sqrt{L}\hat{m}_L)$ converges to the bivariate normal with zero mean and covariance matrix

$$\left(\begin{array}{cc} H^{-2}A & 0\\ 0 & \rho_0^2 B \end{array}\right)$$

as $L \to \infty$, where

$$H = G(0) - 2\sin\left(\frac{\pi}{M}\right) \sum_{k=0}^{M-1} g\left(\frac{2\pi}{M}k + \frac{\pi}{M}\right),$$
$$A = \mathbb{E}R_1^2 \sin^2 \left\langle \Phi_1 \right\rangle, \quad B = \mathbb{E}R_1^2 \cos^2 \left\langle \Phi_1 \right\rangle - G^2(0).$$

We now discuss the assumptions made by these theorems. The assumption that $w_1, \ldots w_L$ are circularly symmetric can be relaxed, but this comes at the expense of making the theorem statements more complicated. If w_i is not circularly symmetric then the distribution of R_i and Φ_i may depend on a_0 and also on the transmitted symbols s_1, \ldots, s_L . In result, the asymptotic variance described in Theorem 2 depends on a_0 and s_1, \ldots, s_L , rather than just ρ_0 . The circularly symmetric assumption may not always hold in practice, but we feel it provides a sensible trade off between simplicity and generality. A key assumption in Theorem 1 is that G(x) is uniquely maximised at x = 0 for $x \in [-\frac{\pi}{M}, \frac{\pi}{M}]$. Although we will not prove it here, this assumption is not only sufficient, but also necessary, for if G(x) is uniquely maximised at some $x \neq 0$ then $\langle \hat{\theta} - \theta_0 \rangle_{\pi} \to x$ almost surely as $L \to \infty$, while if G(x) is not uniquely maximised then $\langle \hat{\theta} - \theta_0 \rangle_{\pi}$ will not converge. One can check that the assumption holds when w_1 is circularly symmetric and normally distributed.

The theorems make statements about $\langle \hat{\theta} - \theta_0 \rangle$ rather than directly on $\hat{\theta} - \theta_0$. This makes practical sense, and to see why let $s'_i = e^{j2\pi k/M} s_i$ be *M*-PSK symbols obtained by rotating s_1, \ldots, s_L by $e^{j2\pi k/M}$ for some integer *k*. Then

$$as_i = \rho e^{j\theta} s_i = \rho e^{j(\theta - 2\pi k/M)} s'_i$$

and so, if $a = \hat{\rho}e^{j\hat{\theta}}$ minimises (3), then so does $\hat{\rho}e^{j(\hat{\theta}-2\pi k/M)}$. Thus, $\hat{\theta} - \theta_0$ should be attributed the same error as $\hat{\theta} - \theta_0 - \frac{2\pi}{M}k$. That is, the error should be computed 'modulo $\frac{2\pi}{M}$ '. It is for this reason that *differential encoding* is often used with noncoherent *M*-PSK detectors.

4. THE GAUSSIAN NOISE CASE

Let the noise sequence $\{w_i\}$ be complex Gaussian with independent real and imaginary parts having zero mean and variance σ^2 . The joint density function of the real and imaginary parts is

$$\frac{1}{2\pi\sigma^2}e^{-\frac{1}{2\sigma^2}(x^2+y^2)}$$

Theorem 1 and 2 hold, and since the distribution of w_1 is circularly symmetric, the distribution of $R_1 e^{j\Phi_1}$ is identical to the distribution of $1 + \frac{1}{\rho_0}w_1$. It can be shown that

$$g(\phi) = \frac{\cos(\phi)}{2\pi} e^{-\frac{1}{2}\kappa^2} + \frac{\Psi(\phi)}{\kappa\sqrt{2\pi}} e^{-\frac{1}{2}\kappa^2\sin^2(\phi)} \left(1 + \kappa^2\cos^2(\phi)\right)$$

where $\kappa = \rho_0 / \sigma$ and

$$\Psi(\phi) = \frac{1}{2} + \frac{1}{2} \operatorname{erf}\left(\frac{\kappa \cos(\phi)}{\sqrt{2}}\right)$$

and $\operatorname{erf}(x) = \frac{2}{\sqrt{\pi}} \int_0^x e^{-t^2} dt$ is the error function. The value of A and B can be efficiently computed by numerical integration using these formula.

5. SIMULATIONS

We present the results of Monte-Carlo simulations with the least squares estimator. In all simulations the noise w_1, \ldots, w_L is independent and identically distributed circularly symmetric and Gaussian with real and imaginary parts having variance σ^2 . Simulations are run with M = 2, 4(BPSK and QPSK) and L = 16,256,4096 with signal to noise ratio SNR $= \frac{\rho_0^2}{2\sigma^2}$ between -20 dB and 20 dB in steps of 1 dB. The amplitude $\rho_0 = 1$ and θ_0 is uniformly distributed on $[-\pi, \pi)$. For each value of signal to noise ratio T = 10000 replications are performed to obtain T estimates $\hat{\rho}_1, \ldots, \hat{\rho}_T$ and $\hat{\theta}_1, \ldots, \hat{\theta}_T$.

Figures 1 and 2 show the sample mean square error (MSE) of the phase estimator $\hat{\theta}$ computed as $\frac{1}{T} \sum_{i=1}^{T} \langle \hat{\theta}_i - \theta_0 \rangle^2$. The dots show the sample MSE of the least square estimator. The dashed line is the sample MSE of the unmodulated carrier estimator (2) that has a priori knowledge of the transmitted symbols s_1, \ldots, s_L . The solid line is the MSE predicted by Theorem 2. The theorem accurately predicts the behaviour of the phase estimator when L is sufficiently large and when the SNR is not too small. As the SNR decreases the variance of the phase estimator approaches that of the uniform distribution on $\left[-\frac{\pi}{M}, \frac{\pi}{M}\right)$ and Theorem 2 does not model this behaviour.

Figures 1 and 2 also display the sample MSE of the phase estimator of Viterbi and Viterbi [1]. This estimator requires the selection of a function F that transforms the amplitude of each sample prior to the final estimation step. They propose several viable alternatives, from which we have chosen F(x) = 1. The sample MSE of the least squares estimator and the Viterbi and Viterbi estimator is similar. The least squares estimator appears slightly more accurate at low SNR.

Figures 3 and 4 show the variance of the amplitude estimator $\hat{\rho}$. The solid line is the asymptotic variance predicted by Theorem 2, the dots are the Monte-Carlo simulations with the least squares estimator, and the dashed line the simulations with the unmodulated carrier estimator. For the least squares estimator each point is computed as $\frac{1}{T} \sum_{i=1}^{T} (\hat{\rho}_i - \rho_0 G(0))^2$. This requires G(0) to be known. In practice G(0) may not be known at the receiver, so Figures 3 and 4 serve to validate the correctness of our asymptotic theory, rather than to suggest the practical performance of the amplitude estimator. When SNR is large G(0) is close to 1 and the bias of the amplitude estimator is small. However, G(0) grows without bound as the variance of the noise increases, so the bias is significant when SNR is small. As indicated in Figures 3 and 4 the variance of the least squares amplitude estimator is smaller than that of the unmodulated carrier. However, due to the bias, the MSE of the least squares amplitude estimator is *not* smaller than that of the unmodulated carrier.

6. CONCLUSION

We have studied the least squares estimator of carrier phase and amplitude from the observation of L noisy M-PSK symbols. The estimator can be computed in $O(L \log L)$ operations using the algorithm of Mackenthun [6], and is the maximum likelihood estimator in the case that the noise is additive white and Gaussian. We showed that the phase estimator $\hat{\theta}$ is strongly consistent and asymptotically normally distributed. However, the amplitude estimator $\hat{\rho}$ is biased, and converges to $G(0)\rho_0$. This bias is large when the signal to noise ratio is small. It would be interesting to investigate methods for correcting this bias.



Fig. 1. Phase mean square error for BPSK



Fig. 3. Amplitude variance for BPSK



Fig. 2. Phase mean square error for QPSK

Fig. 4. Amplitude variance for QPSK

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