Interleaved Direct Bandpass Sampling for Software Defined Radio/Radar Receivers

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Outline

- Receiver Architectures
- Bandpass Sampling Problem
- Interleaved ADC Sampling Model
- Complex Envelope Computation
- Computer Simulations
- Interleaved ADC Calibration
- Conclusions



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Receiver Architectures

• Consider modulated signal

$$x_{c}(t) = \Re\{c_{c}(t)e^{j\Omega_{c}t}\}$$
$$= a_{c}(t)\cos(\Omega_{c}t) - b_{c}(t)\sin(\Omega_{c}t)$$

of $x_c(t)$, where Ω_c = carrier frequency and $c_c(t) = a_c(t) + jb_c(t) =$ complex envelope with bandwidth B/2.

- We seek to compute the sampled sequences $a(n) = a_c(nT'_s)$ and $b(n) = b_c(nT'_s)$ of the in phase and quadrature signal components, where $\Omega'_s > B$.
- Classical heterodyne and homodyne receiver architectures require analog filters and mixers.

Receiver Architectures (cont'd)



Heterodyne Receiver Architecture



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Receiver Architectures (cont'd)



Homodyne Receiver



Receiver Architectures (ct'd



Direct Interleaved Sampling Receiver

Uses a high quality tunable MEMS filter H_S , 2 wideband S/Hs (available from Inphi), but no mixer or phase shifter and fewer analog filters.





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Bandpass Sampling Problem

• Consider a bandpass signal whose spectral support is $[-\Omega_H, -\Omega_L]$ and $[\Omega_L, \Omega_H]$, with occupied bandwidth $B = \Omega_H - \Omega_L$.





Bandpass Sampling Problem (cont'd)

- Due to overlap of copies X₊(j(Ω − kΩ_s)) and X₋(j(Ω − ℓΩ_s)) of the positive and negative components of the spectrum (with k, ℓ integer), it is not always possible to reconstruct x_c(t) from its samples x(n) = x_c(nT_s) if Ω_s > 2B.
- Stronger conditions need to be satisfied. Let $k_M = \lfloor \Omega_H / B \rfloor$. Then Ω_s must satisfy

$$\frac{2\Omega_H}{kB} \le \frac{\Omega_s}{B} \le \frac{2\Omega_L}{(k-1)B}$$

with k integer such that $1 \le k \le k_M$.



Bandpass sampling (cont'd)



Bandpass Sampling (cont'd)

Kohlenberg (1953), Coulson (1995), Lin and Vadyanathan (1998) recognized that x_c(t) can be recovered from the samples x₁(n) = x_c(nT'_s) and x₂(n) = x_c((n - d)T'_s) of two interleaved ADCs, as long as the overall sampling rate Ω_s = 2Ω'_s > 2B. But some offsets, such as d = 1/2, are forbidden.







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Interleaved ADC Sampling Model

• Consider the representation

$$x_c(t) = \Re\{c_c(t)e^{j\Omega_c t}\}\$$

of $x_c(t)$, where Ω_c = carrier frequency and $c_c(t) = a_c(t) + jb_c(t) =$ complex envelope with bandwidth B/2.

• Unlike Kohlenberg and others who considered the problem of recovering $x_c(t)$ from its interleaved samples, we seek to evaluate the sampled complex envelope $c(n) = c_c(nT_s)$ from the interleaved samples, i.e. we jointly sample and demodulate.



Interleaved ADC Sampling Model (cont'd)

• Let $\ell = \operatorname{round}(\Omega_c / \Omega'_s)$ and

so that $-\pi \leq \omega_b < \pi$.

$$\omega_b = \left(\frac{\Omega_c}{\Omega'_s} - \ell\right) 2\pi \;,$$

$$\begin{array}{c|c} & & \ell \text{-th image} \\ & & \text{zone} \\ & & & \\ \hline \end{array} \\ \hline & & & \\ \hline \hline & & & \\ \hline \end{array} \\ \hline & & & \\ \hline \hline \\ \hline & & & \\ \hline \end{array} \\ \hline \end{array} \\ \hline \\ \hline \hline \\ \hline \end{array} \\ \hline \end{array} \\ \hline \end{array}$$



Interleaved ADC Sampling Model (cont'd)

With $\Omega'_s > B$, if $c(n) = c_c(nT'_s)$ = sampled envelope, we have

$$x_1(n) = \Re\{c(n)e^{j\Omega_c T'_s n}\} = \Re\{c(n)e^{j\omega_b n}\}$$

and

$$x_2(n) = \Re\{(f * c)(n)e^{j\Omega_c T'_s(n-d)}\}$$
$$= \Re\{(f * c)(n)e^{j\omega_b(n-d)}e^{-j2\pi\ell d}\}$$

where f(n) = impulse response of the fractional delay filter $F(e^{j\omega}) = e^{-j\omega d}$ for $-\pi < \omega \le \pi$. Let

$$G(e^{j\omega}) = \begin{cases} e^{-j2\pi(\ell+1)} & -\pi \le \omega < -\pi + \omega_b \\ e^{-j2\pi\ell d} & -\pi + \omega_b \le \omega < \pi \end{cases}$$

for $\omega_b \geq 0$, and

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Interleaved ADC Sampling Model (cont'd)

$$G(e^{j\omega}) = \begin{cases} e^{-j2\pi\ell d} & -\pi \le \omega \le \pi + \omega_b \\ e^{-j2\pi(\ell-1)d} & \pi + \omega_b \le \omega < \pi \end{cases}$$

for $\omega_b < 0$. This gives the model shown below, where the goal is to recover c(n) from $x_1(n)$ and $x_2(n)$.







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Complex Envelope Computation

We have

$$\begin{bmatrix} X_1(e^{j\omega}) \\ X_2(e^{j\omega}) \end{bmatrix} = \begin{bmatrix} 1 & 0 \\ 0 & F(e^{j\omega}) \end{bmatrix} \mathbf{M}(e^{j\omega}) \begin{bmatrix} C(e^{j(\omega-\omega_b)}) \\ C^*(e^{-j(\omega+\omega_b)}) \end{bmatrix}$$

where

$$\mathbf{M}(e^{j\omega}) = \frac{1}{2} \begin{bmatrix} 1 & 1 \\ G(e^{j\omega}) & G^*(e^{-j\omega}) \end{bmatrix}$$

and $X_1(e^{j\omega})$, $X_2(e^{j\omega})$ and $C(e^{j\omega}) = \text{DTFTs of } x_1(n)$, $x_2(n)$ and c(n). So $C(e^{j(\omega-\omega_b)})$ can be recovered from $X_1(e^{j\omega})$ and $X_2(e^{j\omega})$ as long as $\mathbf{M}(e^{j\omega})$ is invertible.

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The determinant $D(e^{j\omega})$ of $\mathbf{M}(e^{j\omega})$ is given by

$$D(e^{j\omega}) = \frac{j}{2}\sin(2\pi\ell d)$$

for $0 \leq \omega \leq \pi - |\omega_b|$ and

$$D(e^{j\omega}) = \frac{j}{2} e^{j\pi \operatorname{sgn}(\omega)d} \sin(\pi(2\ell + \operatorname{sgn}(\omega_b))d,$$

for $\pi - |\omega_b| \le \omega < \pi$. Accordingly $D(e^{j\omega})$ is nonzero as long as $\ell \ge 1$ and the offset d is not equal to one of the forbidden values

$$d_m^i = \frac{m}{2\ell} , \ d_q^e = \frac{q}{2\ell + \operatorname{sgn}(\omega_b)}$$

with m and q integers such that $0 \le m \le 2\ell - 1$ and $1 \le q \le 2\ell + \operatorname{sgn}(\omega_b) - 1$.

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The complex envelope c(n) can be computed as shown below, where

$$\begin{bmatrix} H_1(e^{j\omega}) & H_2(e^{j\omega}) \end{bmatrix} = \begin{bmatrix} 1 & 0 \end{bmatrix} \mathbf{M}^{-1}(e^{j\omega}) \begin{bmatrix} 1 & 0 \\ 0 & F^{-1}(e^{j\omega}) \end{bmatrix}$$
$$= \frac{1}{2D(e^{j\omega})} \begin{bmatrix} G^*(e^{-j\omega}) - F^{-1}(e^{j\omega}) \end{bmatrix}.$$





• The filters $H_1(e^{j\omega})$ and $H_2(e^{j\omega})$ are noncausal IIR with impulse responses

$$\Re\{h_1(n)\} = \delta(n) , \ \Re\{h_2(n)\} = 0$$

$$\Im\{h_1(n)\} = -\cot(\pi(2\ell + \operatorname{sgn}(\omega_b))d)\delta(n) + \left(\cot(\pi(2\ell + \operatorname{sgn}(\omega_b))d) - \cot(2\pi\ell d)\right)\frac{\sin((\pi - |\omega_b|)n)}{\pi n}$$

$$\Im\{h_2(n)\} = \frac{\sin((\pi - |\omega_b|)(n+d))}{\pi \sin(2\pi\ell d)(n+d)} - \frac{\sin((\pi - |\omega_b|)(n+d) - \pi d)}{\pi \sin(\pi(2\ell + \operatorname{sgn}(\omega_b)d)(n+d))}$$

They can be approximated by causal FIR filters of order M=2L by shifting the responses by *L* and applying a Kaiser window of order *M*.
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• The quadrature sampling scheme proposed by Brown (1979) requires $\Omega'_s = \Omega_c/\ell \ (\omega_b = 0)$ and $d = \frac{1}{4\ell} + \frac{q}{\ell}$ with q integer. In this case

$$x_1(n) = a_c(nT'_s)$$
, $x_2(n) = b_c((n-d)T'_s)$

are the sampled in-phase and quadrature components of the complex envelope

- Impractical for software defined radio since it ties the sampling frequency to the carrier frequency.
- The proposed scheme requires only Ω_s > 2B and ℓ ≥ 1. It is well adapted to software defined radio: if Ω_c changes, ℓ and ω_b change and the filters h₁(n) and h₂(n) can be recomputed by windowing their closed form expressions.





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Computer Simulations

• The bandpass signal $x_c(t)$ has envelope

 $c_{c}(t) = 2\cos(400 \times 10^{6} \times 2\pi t) + j[\sin(400 \times 10^{6} \times 2\pi t) + \cos(175 \times 10^{6} \times 2\pi t)]$

with bandwith $(B/2)/2\pi = 400$ MHz, and carrier frequency $F_c = \Omega_c/(2\pi) = 5.15$ GHz.

- The sub-ADC sampling frequency $F_s' = \Omega_s'/(2\pi) = 1$ GHz is above B = 800MHz.
- $F_c = 5F'_s + 150$, so $\ell = 5$ and $\omega_b = 0.3\pi$.
- The sampled envelope

$$c(n) = \frac{3}{2}e^{j0.8\pi n} + \frac{1}{2}e^{-j0.8\pi n} + \frac{j}{2}[e^{j0.35\pi n} + e^{-j0.35\pi n}]$$

has tones at $\pm 0.8\pi$ and $\pm 0.35\pi$.

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Computer Simulations (cont'd)



- Timing offset d = 0.425, between $d_4^i = 0.4$ and $d_5^e = 0.454$.
- Additive noises with SNR =62dB added to x₁(n) and x₂(n) to model thermal, quantization noises
- Kaiser windows of order M = 60 and β = 6 used to approximate filters H₁ and H₂.
- MSE= -53.54dB, SFDR = 65dB.
- Secondary tones = residual components of $e^{-j2\omega_b n}c^*(n)$.



Computer Simulations (cont'd)



- Timing offset d = dⁱ₄ + 0.001
 close to a forbidden offset.
- D(e^{jω}) almost singular over [-0.7π, 0.7π], so filter gains inaccurate over this band.
- Translated to $[-\pi, 0.4\pi]$ by $e^{-j0.3\pi n}$ demodulation.
- MSE = -47.27dB, SFDR =55dB.







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Interleaved ADC Calibration

- The effect of timing-skew mismatches δ = d − d₀ with d = true mismatch, d₀ = nominal mismatch, grows as increases. Due to the 2πℓd phase shift of G(e^{jω}).
- ADC_1 and ADC_2 cannot share the sample S/H since $d \neq 1/2 \Rightarrow$ mismatches cannot be avoided.
- TIADC calibration is therefore needed. We propose a blind calibration method under the assumption that the signal is oversampled with oversampling ratio $\alpha = 1 2B/\Omega_s$.



Interleaved ADC Calibration

• If $\gamma = g - 1 =$ relative gain mismatch, if

$$\boldsymbol{ heta} = \left[egin{array}{c} g \\ d \end{array}
ight] \ , \ \boldsymbol{ heta}_0 = \left[egin{array}{c} 1 \\ d_0 \end{array}
ight]$$

denote the true and nominal parameter vectors (g = relative channel gain), for small mismatches we use first-order expansions

$$H_1(e^{j\omega}, d) = H_{10}(e^{j\omega}) + \delta H_{11}(e^{j\omega})$$
$$H_2(e^{j\omega}, \boldsymbol{\theta}) = (1 - \gamma)[H_{10}(e^{j\omega}) + \delta H_{11}(e^{j\omega})].$$

 Due to oversampling, C(e^{jω}) = 0 over J = [(1 - α)π, (1 + α)π] mod (2π), so r(n, θ) = c(n)e^{jω_bn} has no power in the band J + ω_b mod (2π).



Interleaved ADC Calibration

• let $h_{BP}(n)$ = impulse response of bandpass filter

$$H_{\mathrm{BP}}(e^{j\omega}) = \begin{cases} 1 & \omega \in I \\ 0 & \text{otherwise} \end{cases}.$$

Then the error signal

$$e(n, \hat{\theta}) = h_{\mathrm{BP}}(n) * r(n, \hat{\theta})$$

is zero for $\hat{\theta} = \theta$, nonzero otherwise.

• The blind estimation algorithm minimizes adaptively

$$J(\hat{\theta}) = E[|e(n, \hat{\theta})|^2]$$

by using the stochastic gradient scheme

$$\hat{\boldsymbol{\theta}}(n+1) = \hat{\boldsymbol{\theta}}(n) - \mu \Re\{e^*(n, \hat{\boldsymbol{\theta}}(n) \nabla_{\hat{\boldsymbol{\theta}}} e(n, \hat{\boldsymbol{\theta}}(n))\}.$$

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- Mismatches: $\gamma = 10^{-2}$, $\delta = -0.25 \times 10^{-2}$.
- $I = [-0.9\pi, -0.5\pi]$, FIR approximation of $H_{\rm BP}$ with Kaiser window of order M = 80.
- Step sizes $\mu_{\gamma} = 10^{-3}$, $\mu_{\delta} = 10^{-5}$.
- $L = 5 \times 10^4$ samples. Final estimates $\hat{\gamma}(L) = 0.77 \times 10^{-2}, \hat{\delta} = -0.27 \times 10^{-2}.$





Before calibration: MSE=-17.8dB, After calibration: MSE=-36.51dB, SFDR=25dB SFDR=43dB.

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Carrier Freq.	2.15GHz	3.15Gz	4.15GHz	5.15 GHz
l	2	3	4	5
MSE, Before (dB)	-25.51	-22.24	-19.76	-17.79
MSE, After (dB)	-50.89	-46.01	-40.97	-36.52
$\hat{\gamma}(L)(\times 10^{-3})$	10	9.7	8.8	7.7
$\hat{\delta}(L) \ (imes 10^{-3})$	2.5	2.6	2.6	2.7

The performance of the envelope reconstruction scheme before and after calibration degrades as ℓ increases. Reason: the first order expansion of correction filters H_1 and H_2 becomes inaccurate. As soon as g and d are estimated, the exact filters H_1 and H_2 should be used instead of their first order approximations.





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Conclusions

- A direct complex envelope sampling scheme for bandpass signals with nonuniformly interleaved ADCs has been proposed. In combination with tunable RF band selection filters and high bandwidth S/H circuits, provides an approach to software defined radio/radar.
- TIADC calibration is required due to the sensitivity of the sampling system to timing skew mismatches.
- The proposed sampling and blind calibration system work well for small to moderate values of l (Ω_c/B). The calibration can be improved for large values of l by iterating the calibration using the most recent θ estimate as reference in the first-order expansions of H₁(e^{jω}, d) and H₂(e^{jω}, θ).





Thank you!

