Study on possible error in signal amplitude and time of arrival reconstruction

Dawid Pietruch

AGH University of Krakow, Faculty of Physics and Applied Computer Science

pietruch@agh.edu.pl

26.09.2024

This research was funded by the National Science Centre, Poland, under the grant no. 2021/43/B/ST2/01107

Motivation

This study was performed to understand possible errors in the reconstruction of amplitude and time of arrival using fit to raw data and deconvolution method. Steps:

- Creating generator of virtual pulses which mimic raw data with given amplitude, TOA and level of noise
- Running deconvolution reconstruction on this data
- Comparison of methods which are responsible for choosing best set of samples for deconvolution
- Running fitting function for generated data

Two approaches to reconstruct amplitude of signal and time of arrival

Deconvolution method

$$d_{i} = v_{i} - 2e^{-\frac{T_{smp}}{\tau_{sh}}}v_{i-1} + e^{-\frac{2T_{smp}}{\tau_{sh}}}v_{i-2}$$

$$t_0 = rac{rac{d_2}{d_1} T_{smp}}{rac{d_2}{d_1} + e^{-rac{T_{smp}}{ au_{sh}}}}$$

$$V(t) = egin{cases} b, & ext{for } t < t_0. \ lpha \left(rac{t-t_0}{ au_{sh}}
ight) e^{-rac{t-t_0}{ au_{sh}}} + b, & ext{for } t \geq t_0. \end{cases}$$

- pedestal subtraction
- pulse detection
- common mode subtraction
- signal amplitude and TOA reconstruction by applying fitting procedure

$$A = (d_1 + d_2) \left[\frac{\tau_{sh}}{T_{smp}} e^{\frac{T_{smp} - \tau_{sh}}{\tau_{sh}}} \right] \frac{e^{\frac{-t_0}{\tau_{sh}}}}{1 - \frac{t_0}{T_{smp}} \left(1 - e^{-\frac{T_{smp}}{\tau_{sh}}}\right)}$$

- pedestal subtraction
- pulse detection
- common mode subtraction
- calculating FIR samples (1 or 2 nonzero samples in ideal case)
- pair of FIR samples selection
- time of arrival calculation
- amplitude of signal calculation (based on selected FIR samples and calculated TOA)

Virtual pulse generator

Virtual pulse generator produce ideal CR-RC shaping pulses with additional Gaussian noise $n(\mu, \sigma)$, pulse start t_0 , amplitude α and pedestal b

$$V(t) = \begin{cases} b + n(\mu, \sigma), & \text{for } t < t_0. \\ \alpha\left(\frac{t - t_0}{\tau_{sh}}\right) e^{-\frac{t - t_0}{\tau_{sh}}} + b + n(\mu, \sigma), & \text{for } t \ge t_0. \end{cases}$$

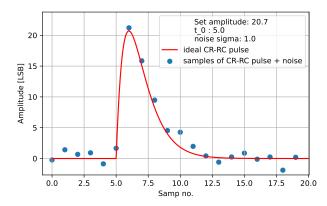


Fig. 1: An example of generated pulse and theoretical noiseless pulse $\textcircled{\sc l}$

Animations

< □ > < ⑦ > < 言 > < 言 > こ ● < ○ へ () 5/31

Fit performance - signal amplitude

Fitting function was checked on generated data without digitisation and additional noise σ =0 LSB.

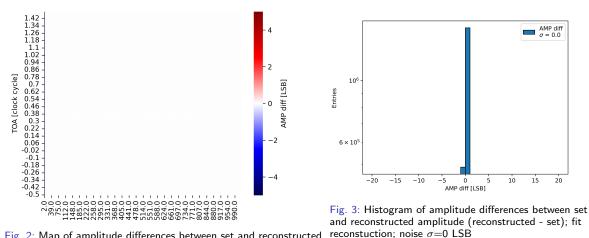


Fig. 2: Map of amplitude differences between set and reconstructed r amplitude (reconstructed - set); fit reconstruction; noise $\sigma=0$ LSB

Fit performance - time of arrival

Fitting function was checked on generated data without digitisation and additional noise σ =0 LSB.

0.0100 1.42 -1.34 TOA diff 1.26 0.0075 $\sigma = 0.0$ 1.18 1.1 10^{6} 1.02 0.0050 0.94 0.86 cycle] 0.78 0.0025 0.7 0.62 Entries TOA [clock 0.54 - 0.0000 0.46 0.38 0.3 105 -0.00250.22 0.14 diff 0.06 -0.02 -0.0050 O -0.1 -0.18 -0.26 -0.0075-0.34 -2.0-1.5-1.0-0.50.0 0.5 1.0 1.5 2.0 -0.42TOA [clock cycle] -0.5 -0.0100 00000000 Fig. 5: Histogram of TOA differences between set and യ്ഥ് -i - 4 0 reconstructed TOA (reconstructed - set); fit

Fig. 4: Map of TOA differences between set and reconstructed TOA (reconstructed - set); fit reconstruction; noise σ =0 LSB

reconstruction: noise $\sigma=0$ LSB

Digitisation error - signal amplitude

Influence of digitisation on amplitude reconstruction, noise σ =0 LSB.

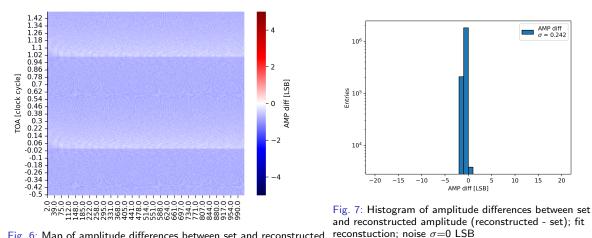


Fig. 6: Map of amplitude differences between set and reconstructed \neg amplitude (reconstructed \neg set); fit reconstruction; noise $\sigma=0$ LSB

LSB

Digitisation error - signal amplitude

Influence of digitisation on amplitude reconstruction, noise $\sigma=0$ LSB.

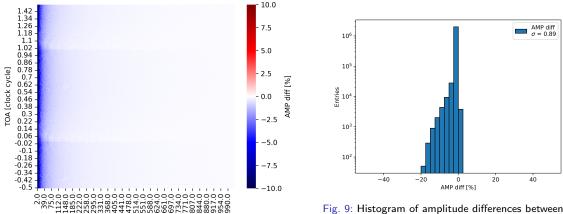


Fig. 8: Map of amplitude differences between set and reconstructed amplitude (reconstructed/set)-100%; fit reconstruction; noise $\sigma=0$

Fig. 9: Histogram of amplitude differences between set and reconstructed amplitude (reconstructed/set)-100%; fit reconstuction: noise $\sigma=0$ LSB

Digitisation error - time of arrival

Influence of digitisation on time of arrival reconstruction, noise $\sigma=0$ LSB.

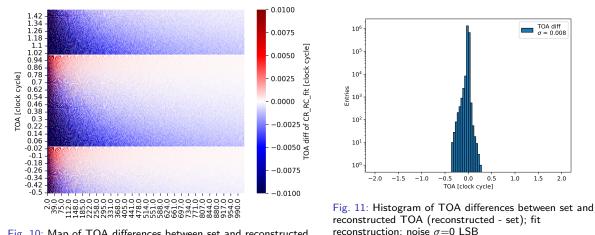


Fig. 10: Map of TOA differences between set and reconstructed TOA (reconstructed - set); fit reconstruction; noise σ =0 LSB

20

Noise influence - signal amplitude

To observe influence of noise on reconstruction of signal amplitude additional Gaussian noise were added to the samples; $\sigma = 1.0$ LSB.

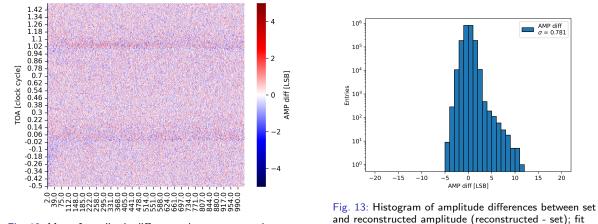


Fig. 12: Map of amplitude differences between set and reconstructed amplitude (reconstructed - set); fit reconstruction; noise $\sigma = 1.0$ LSB

reconstruction; noise σ =1.0 LSB

TOA diff

 $\sigma = 0.02$

2.0

Noise influence - time of arrival

To observe influence of noise on reconstruction of TOA additional Gaussian noise were added to the samples; $\sigma = 1.0$ LSB.

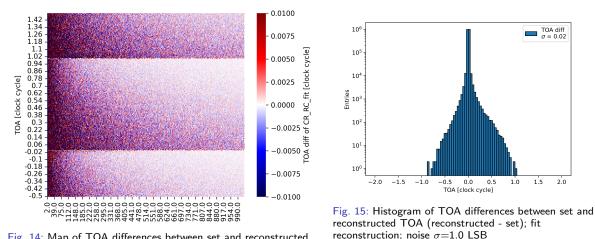


Fig. 14: Map of TOA differences between set and reconstructed TOA (reconstructed - set); fit reconstruction; noise σ =1.0 LSB



1.0

1.5

20

Noise and digitisation influence - signal amplitude

To observe influence of noise on reconstruction of signal amplitude additional Gaussian noise were added to the samples and digitisation was performed; $\sigma = 1.0$ LSB.

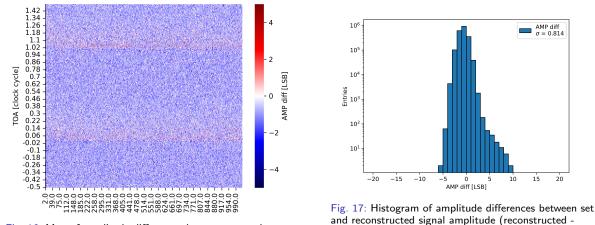


Fig. 16: Map of amplitude differences between set and reconstructed amplitude (reconstructed - set); fit reconstruction; noise $\sigma = 1 \text{ LSB}$

set); fit reconstruction; noise σ =1.0 LSB

Noise and digitisation influence - signal amplitude

To observe influence of noise on reconstruction of signal amplitude additional Gaussian noise were added to the samples and digitisation was performed; σ =1.0 LSB.

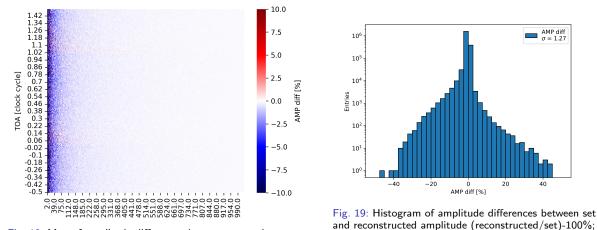


Fig. 18: Map of amplitude differences between set and reconstructed amplitude (reconstructed/set)-100%; fit reconstruction; noise σ =1.0 LSB

fit reconstruction; noise $\sigma = 1.0$ LSB

TOA diff

 $\sigma = 0.019$

2.0

1.5

Noise and digitisation influence - time of arrival

To observe influence of noise on reconstruction of TOA additional Gaussian noise were added to the samples and digitisation was performed; $\sigma = 1.0$ LSB.

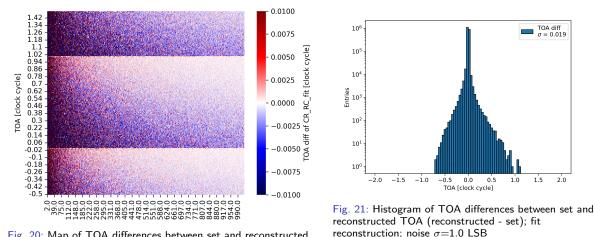


Fig. 20: Map of TOA differences between set and reconstructed TOA (reconstructed - set); fit reconstruction; noise σ =1.0 LSB

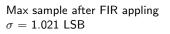
Which FIR samples should be used for deconvolution?

Different methods of determining the best pair of samples for deconvolution were checked.

First ADC sample above treshold

tresnoid

 $\sigma = 1.089 \; \mathrm{LSB}$



 $\begin{array}{l} {\rm Max} \ {\rm ADC} \ {\rm sample} \\ \sigma = 1.019 \ {\rm LSB} \end{array}$

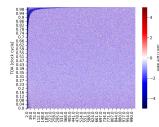


Fig. 22: Map of amplitude differences between set and reconstructed signal amplitude (reconstructed - set); fit reconstruction; noise σ =1.0 LSB

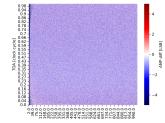


Fig. 23: Map of amplitude differences between set and reconstructed signal amplitude (reconstructed - set); fit reconstruction; noise σ =1.0 LSB

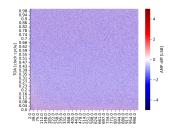


Fig. 24: Map of amplitude differences between set and reconstructed signal amplitude (reconstructed - set); fit reconstruction; noise σ =1.0 LSB

Max sample after FIR appling

Which FIR samples should be used for deconvolution?

Different methods of determining the best pair of samples for deconvolution were checked.

 $\sigma = 1.021$ LSB

First ADC sample above treshold

 $\sigma = 1.089 \text{ LSB}$

AMP diff a = 1.08910⁴ 105 10 Entries 103 107 10) -20 -15 -10 -5 10 15 20 AMP diff [LSB]

Fig. 25: Histogram of amplitude differences between set and reconstructed signal amplitude (reconstructed - set); deconvolution reconstruction; noise σ =1.0 LSB

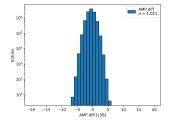


Fig. 26: Histogram of amplitude differences between set and reconstructed signal amplitude (reconstructed - set); deconvolution reconstruction; noise σ =1.0 LSB

Max ADC sample $\sigma = 1.019$ LSB

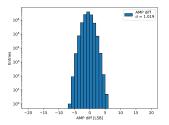


Fig. 27: Histogram of amplitude differences between set and reconstructed signal amplitude (reconstructed - set); deconvolution reconstruction; noise σ =1.0 LSB

<u>e</u>

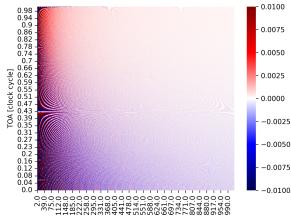
[cloch

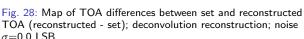
maximum

Ø

Deconvolution digitisation influence - time of arrival

digitisation influence on deconvolution time o arrival reconstruction; ; σ =0.0 LSB.





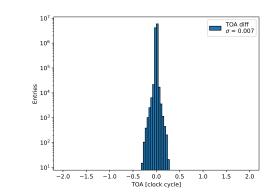


Fig. 29: Histogram of TOA differences between set and reconstructed TOA (reconstructed - set); deconvolution reconstruction; noise σ =0.0 LSB

Deconvolution digitisation influence - signal amplitude

digitisation influence on deconvolution amplitude reconstruction; σ =0.0 LSB.

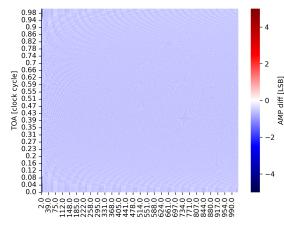


Fig. 30: Map of amplitude differences between set and reconstructed amplitude (reconstructed/set); deconvolution reconstruction; noise σ =0.0 LSB

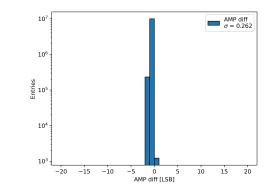


Fig. 31: Histogram of amplitude differences between set and reconstructed amplitude (reconstructed/set); deconvolution reconstruction; noise σ =0.0 LSB

AMP diff

40

 $\sigma = 0.743$

Deconvolution digitisation influence - signal amplitude

digitisation influence on deconvolution amplitude reconstruction; ; σ =0.0 LSB.

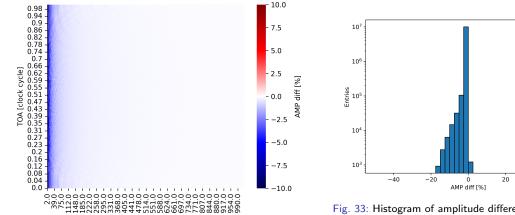


Fig. 32: Map of amplitude differences between set and reconstructed amplitude (reconstructed/set)-100%; deconvolution reconstruction; noise σ =0.0 LSB

Fig. 33: Histogram of amplitude differences between set and reconstructed amplitude (reconstructed/set)-100%; deconvolution reconstruction; noise σ =0.0 LSB

TOA diff

 $\sigma = 0.024$

2.0

э

Deconvolution noise and digitisation influence - time of arrival

To observe influence of noise on reconstruction of TOA additional Gaussian noise were added to the samples and digitisation was performed; $\sigma = 1.0$ LSB.

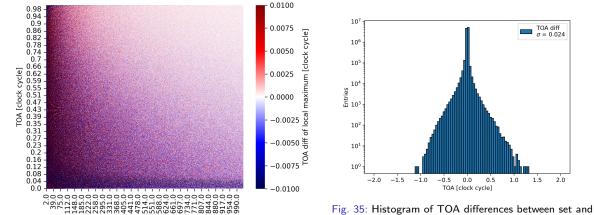


Fig. 34: Map of TOA differences between set and reconstructed TOA (reconstructed - set); deconvolution reconstruction; noise $\sigma = 1.0 \text{ LSB}$

reconstructed TOA (reconstructed - set); deconvolution reconstruction; noise σ =1.0 LSB

・ロッ ・ 一 マ ・ コ ・ ・ 日 ・

Deconvolution noise and digitisation influence - signal amplitude

To observe influence of noise on reconstruction of signal amplitude additional Gaussian noise were added to the samples and digitisation was performed: $\sigma = 1.0$ LSB.

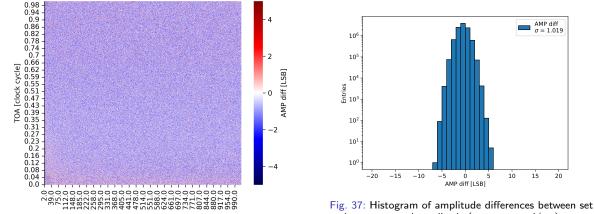


Fig. 36: Map of amplitude differences between set and reconstructed amplitude (reconstructed/set); deconvolution reconstruction; noise $\sigma = 1.0$ LSB

・ロト ・ 一 ト ・ ヨ ト ・ 日 ト

20

э

Deconvolution noise and digitisation influence - signal amplitude

To observe influence of noise on reconstruction of signal amplitude additional Gaussian noise were added to the samples and digitisation was performed; σ =1.0 LSB.

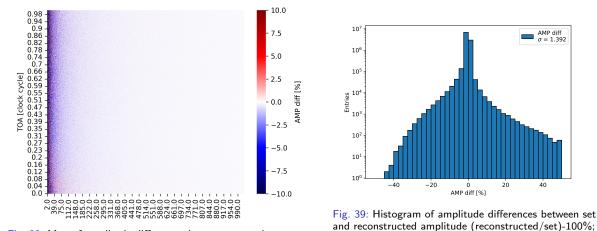


Fig. 38: Map of amplitude differences between set and reconstructed amplitude (reconstructed/set)-100%; deconvolution reconstruction; noise $\sigma{=}1.0$ LSB

deconvolution reconstruction; noise σ =1.0 LSB

Thank you for attention

Vormalized amplitude [-]

Deconvolution method

In our front-end we uses CR-RC shaping, for which amplitude response over time can be written as formula below:

$$V(t) = \frac{q_{in}}{C_{feed}} \frac{t}{\tau_{sh}} e^{-\frac{t}{\tau_{sh}}}$$
(1)

If we include time before pulse, non-zero pulse start time t_0 , amplitude α and pedestal b, equation 1 is transformed into:

$$V(t) = \begin{cases} b, & \text{for } t < t_0. \\ \alpha \left(\frac{t-t_0}{\tau_{sh}} \right) e^{-\frac{t-t_0}{\tau_{sh}}} + b, & \text{for } t \ge t_0. \end{cases}$$
(2)

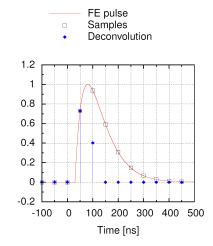


Fig. 40: Example of asynchronous sampling with two nonzero filter output samples at $t_0 = 30$ ns

Our output pulses are a convolution of its impulse response with the sensor's current signal (deposited charge), in order to find the input signal, we can use a procedure inverse to the convolution called deconvolution. In our system, this procedure is performed digitally by a digital filter. Output sample s_k of simplest FIR(Finite Impulse Response):

$$s_k = \sum_{i=0}^{N-1} w_i v_{k-i}$$
 (3)

 w_i - weight associated with input sample v_{k-i}

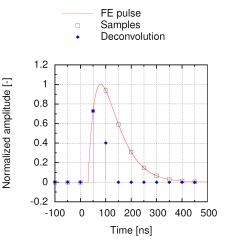


Fig. 41: Example of asynchronous sampling with two nonzero filter output samples at $t_0 = 30$ ns

To calculate amplitude of pulse we have to use some mathematical tools. Let start from front-end response $V_{sh}(s)$ in a Laplace domian can be expressed as:

$$V_{sh}(s) = \frac{1}{s}H(s) = \frac{1}{\tau_{sh}}\frac{1}{\left(s + \frac{1}{\tau_{sh}}\right)^2} \qquad (4)$$

H(s) - transform function of the CR-RC shaper.

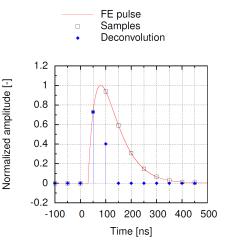


Fig. 42: Example of asynchronous sampling with two nonzero filter output samples at $t_0 = 30$ ns

Next step is to transform from continuous domain s to the discrete domain z using the Z transform. After that and we achieved discrete transform function D(z):

$$D(z) = z^{2} - 2e^{-\frac{T_{smp}}{\tau_{sh}}}z + e^{-\frac{2T_{smp}}{\tau_{sh}}}$$
(5)

Since z^2 represents the sample which will be received after 2 sampling periods, we can just multiplied by z^{-2} delaying all samples by two periods.

$$D(z) = 1 - 2e^{-\frac{T_{smp}}{\tau_{sh}}} z^{-1} + e^{-\frac{2T_{smp}}{\tau_{sh}}} z^{-2}$$
 (6)

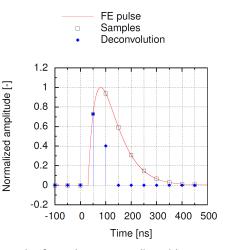


Fig. 43: Example of asynchronous sampling with two nonzero filter output samples at $t_0 = 30$ ns

Output sample value d_i , obtained at time $i \cdot T_{smp}$ can be expressed as:

$$d_{i} = v_{i} - 2e^{-\frac{T_{smp}}{\tau_{sh}}}v_{i-1} + e^{-\frac{2T_{smp}}{\tau_{sh}}}v_{i-2}$$
(7)

where v_i is the shaper output:

$$v_i = V(i \cdot T_{smp}) \tag{8}$$

If we calculate subsequent FIR output samples for CR-RC asynchronous shaper we can notice that filter produces only tow non zero samples or one in synchronous case.

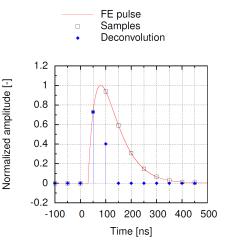


Fig. 44: Example of asynchronous sampling with two nonzero filter output samples at $t_0 = 30$ ns

Ratio between tow non-zero filter samples after reduction is given by:

$$\frac{d_2}{d_1} = \frac{t_0}{T_{smp} - t_0} e^{-\frac{T_{smp}}{\tau_{sh}}}$$
(9)

This ratio enable to calculate pulse starting time (TOA) which is necessary for amplitude reconstruction.

$$t_0 = \frac{\frac{d_2}{d_1} T_{smp}}{\frac{d_2}{d_1} + e^{-\frac{T_{smp}}{\tau_{sh}}}}$$
(10)

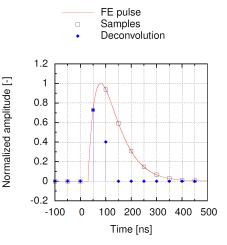


Fig. 45: Example of asynchronous sampling with two nonzero filter output samples at $t_0 = 30$ ns

Sum of two non-zero filter samples after reduction can be expressed as:

$$d_1 + d_2 = \frac{A}{\tau_{sh}} e^{-\frac{T_{smp} - t_0 - \tau_{sh}}{\tau_{sh}}} \left[T_{smp} - t_0 \left(1 - e^{-\frac{T_{smp}}{\tau_{sh}}} \right) \right]$$
(11)

This sum enable to calculate pulse amplitude A

$$A = (d_1 + d_2) \left[\frac{\tau_{sh}}{T_{smp}} e^{\frac{T_{smp} - \tau_{sh}}{\tau_{sh}}} \right] \frac{e^{\frac{-t_0}{\tau_{sh}}}}{1 - \frac{t_0}{T_{smp}} \left(1 - e^{-\frac{T_{smp}}{\tau_{sh}}}\right)}$$
(12)

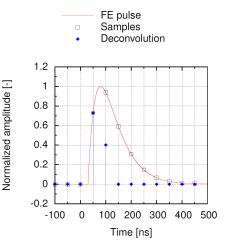


Fig. 46: Example of asynchronous sampling with two nonzero filter output samples at $t_0 = 30$ ns