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Manual | EN

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TwinCAT 3 | Filter Designer
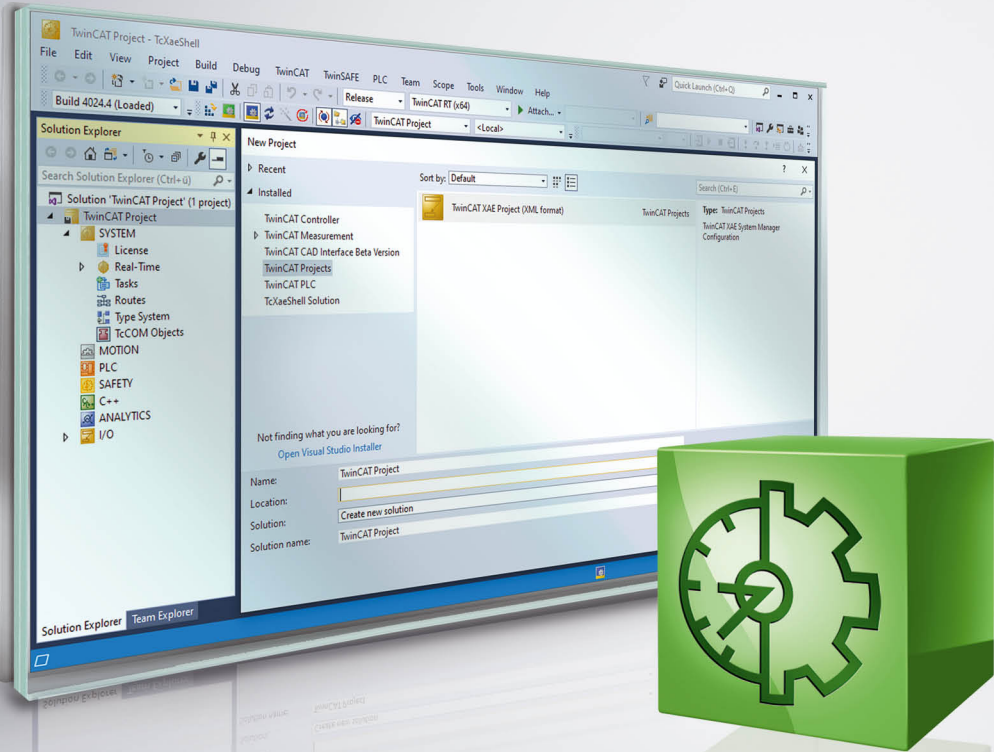
Inhaltsverzeichnis

1 Foreword

1.1 Notes on the documentation

This description is only intended for the use of trained specialists in control and automation engineering who are familiar with applicable national standards.
It is essential that the documentation and the following notes and explanations are followed when installing and commissioning the components.
It is the duty of the technical personnel to use the documentation published at the respective time of each installation and commissioning.

The responsible staff must ensure that the application or use of the products described satisfy all the requirements for safety, including all the relevant laws, regulations, guidelines and standards.

Disclaimer

The documentation has been prepared with care. The products described are, however, constantly under development.
We reserve the right to revise and change the documentation at any time and without prior announcement.
No claims for the modification of products that have already been supplied may be made on the basis of the data, diagrams and descriptions in this documentation.

Trademarks

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Patent Pending

The EtherCAT Technology is covered, including but not limited to the following patent applications and patents:
EP1590927, EP1789857, EP1456722, EP2137893, DE102015105702
with corresponding applications or registrations in various other countries.

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1.2   Safety instructions

Safety regulations

Please note the following safety instructions and explanations!
Product-specific safety instructions can be found on following pages or in the areas mounting, wiring, commissioning etc.

Exclusion of liability

All the components are supplied in particular hardware and software configurations appropriate for the application. Modifications to hardware or software configurations other than those described in the documentation are not permitted, and nullify the liability of Beckhoff Automation GmbH & Co. KG.

Personnel qualification

This description is only intended for trained specialists in control, automation and drive engineering who are familiar with the applicable national standards.

Description of symbols

In this documentation the following symbols are used with an accompanying safety instruction or note. The safety instructions must be read carefully and followed without fail!

⚠ DANGER
Serious risk of injury!
Failure to follow the safety instructions associated with this symbol directly endangers the life and health of persons.

⚠ WARNING
Risk of injury!
Failure to follow the safety instructions associated with this symbol endangers the life and health of persons.

⚠ CAUTION
Personal injuries!
Failure to follow the safety instructions associated with this symbol can lead to injuries to persons.

NOTE
Damage to the environment or devices
Failure to follow the instructions associated with this symbol can lead to damage to the environment or equipment.

● Tip or pointer

i   This symbol indicates information that contributes to better understanding.

2 Overview

The TwinCAT Filter Designer belongs to the TwinCAT Measurement product family. The tool enables the simple design of digital filters. The coefficients determined can be conveniently downloaded for the associated Filter PLC library (TF3680) or into the EtherCAT measurement modules (ELM3xxx).

The Filter Designer is integrated seamlessly into the existing TwinCAT development environment and can be created in a measurement project. You have the option to individualize the filter characteristics graphically with the mouse or text-based in the filter settings.

For large measurement campaigns in which, for example, several ELM modules are to operate with the same filter coefficients on different channels, the TwinCAT Filter Designer offers multi-download strategies in order to simplify the engineering still further.

3 Installation

3.1 System requirements

The following requirements must be satisfied for trouble-free functioning of the TwinCAT Filter Designer.

Supported operating systems

Windows 7, Windows 8.1 and Windows 10

TwinCAT

Minimum is TwinCAT 3.1 Build 4022.29

.NET Framework

Version 4.6.2

Visual Studio® development environment

- Visual Studio® 2015
- Visual Studio® 2017
- TwinCAT XAE Shell

It is basically sufficient to use the Visual Studio® Shell. The "full" setup includes a shell in the form of the TwinCAT XAE Shell. The "update" shell only provides an update of the Filter Designer sources.

3.2 Downloading the setup file

The TwinCAT Measurement Setup can be downloaded from the Beckhoff website. This is the latest version of various measurement products such as Scope View, Scope Server, Bode Plot, Filter Designer etc.. The partly different range of functions can only be activated via the licenses. To download the setup file, perform the following steps:

1. Start a web browser of your choice and go to the Beckhoff website at www.beckhoff.com.
2. In the structure tree of the Web page, navigate to the Automation node and select the relevant product. For example: > TwinCAT 3 > TE1xxx | TC3 Engineering > TE13xx | TC3 Scope View Professional.
3. Click on the Download link in order to place the software in the shopping cart. Then click on **Start download**.
4. (Optional) Transfer the downloaded file to the TwinCAT runtime system you wish to install the product on.
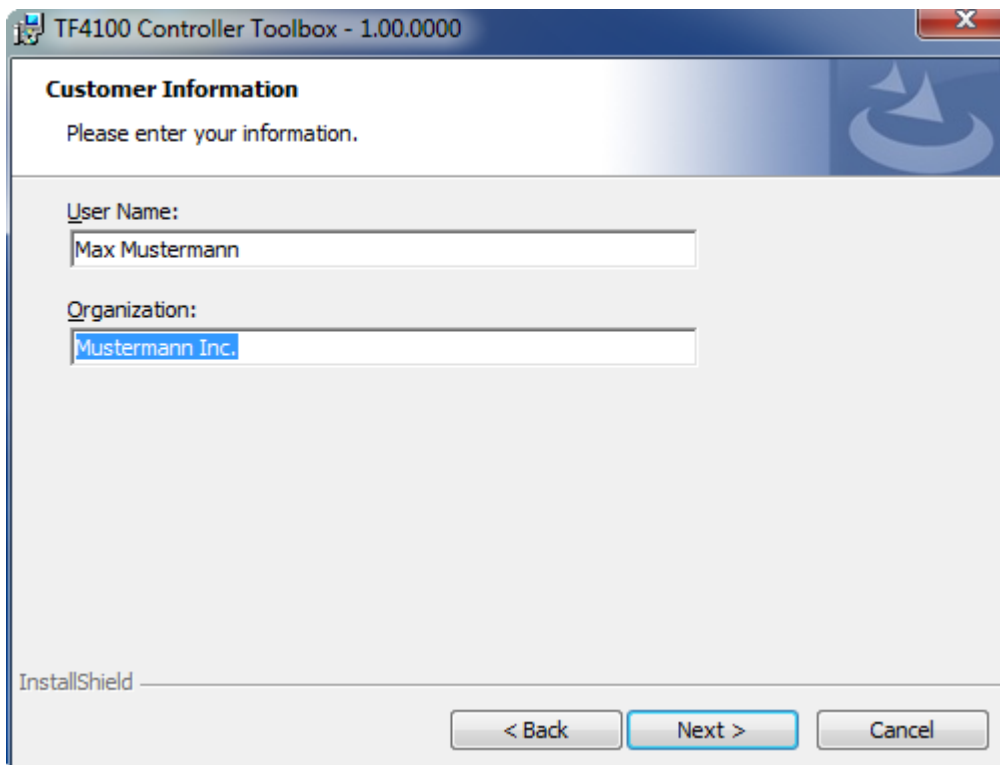
3.3 Installation

The following section describes how to install the TwinCAT 3 Function for Windows-based operating systems.

✓ The TwinCAT 3 Function setup file was downloaded from the Beckhoff website.

1. Run the setup file as administrator. To do this, select the command **Run as administrator** in the context menu of the file.
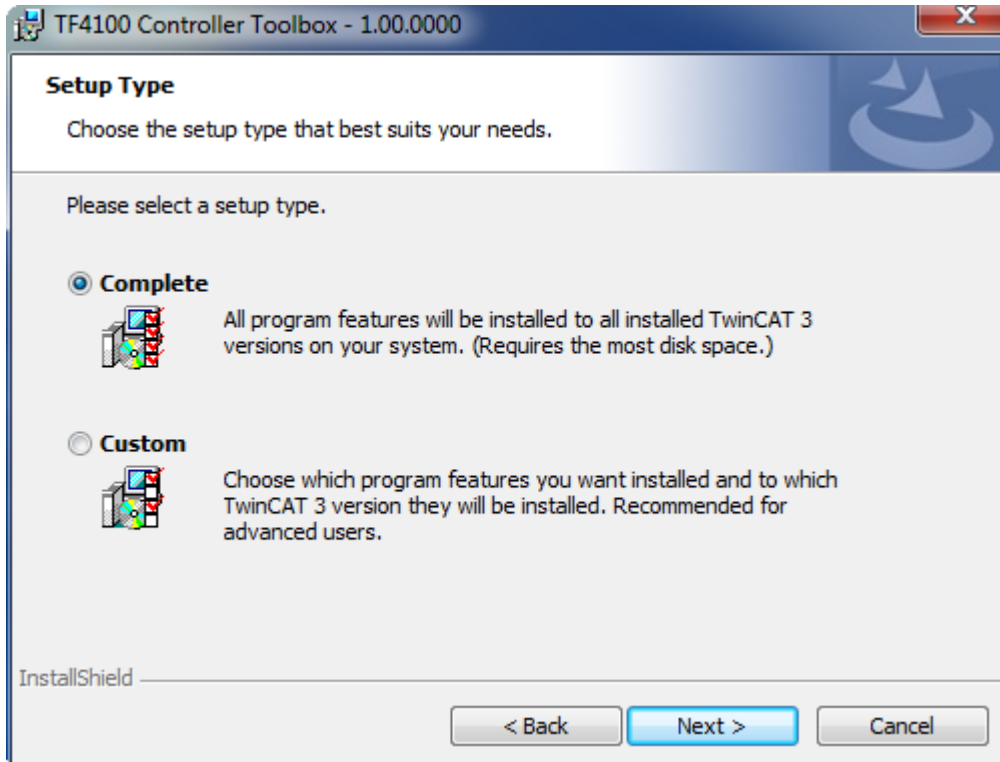   ⇨ The installation dialog opens.

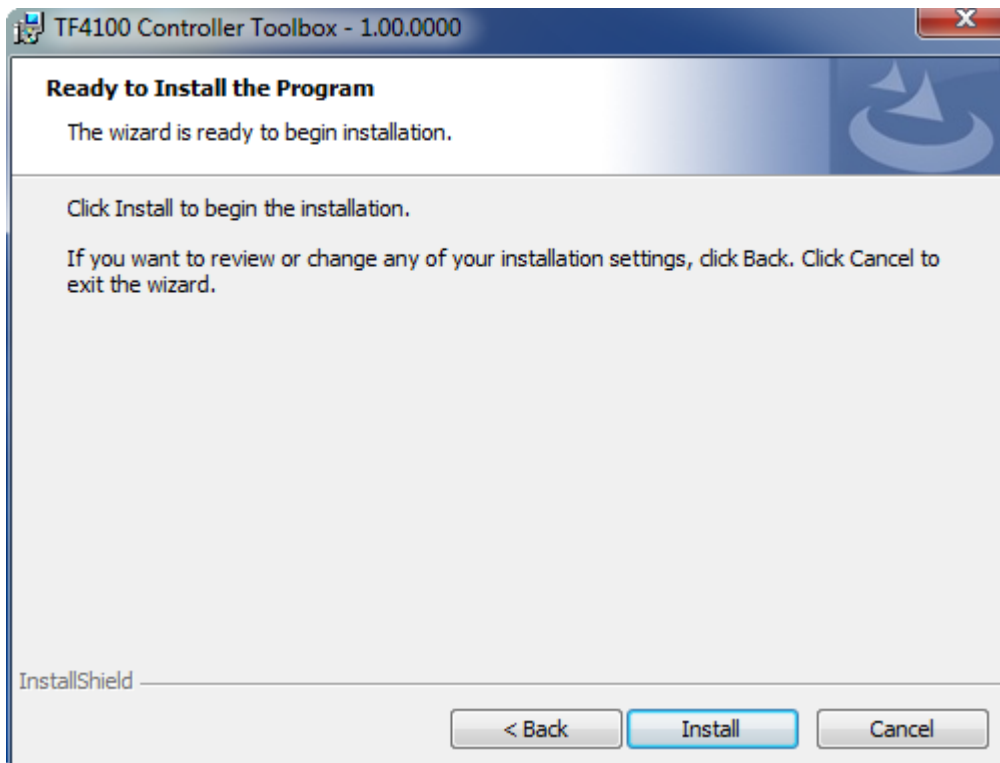2. Accept the end user licensing agreement and click **Next**.

3. Enter your user data.

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4. If you want to install the full version of the TwinCAT 3 Function, select **Complete** as installation type. If you want to install the TwinCAT 3 Function components separately, select **Custom**.
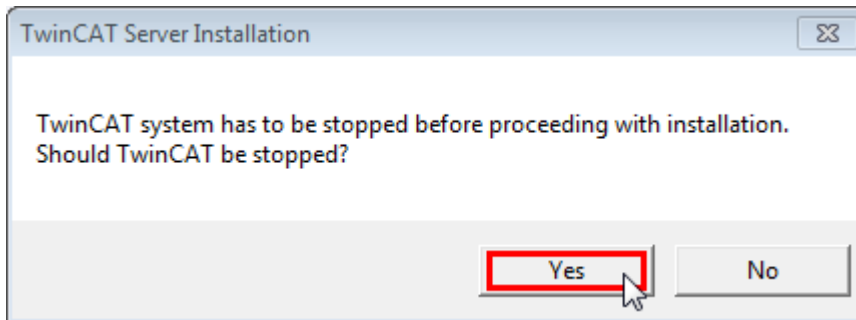

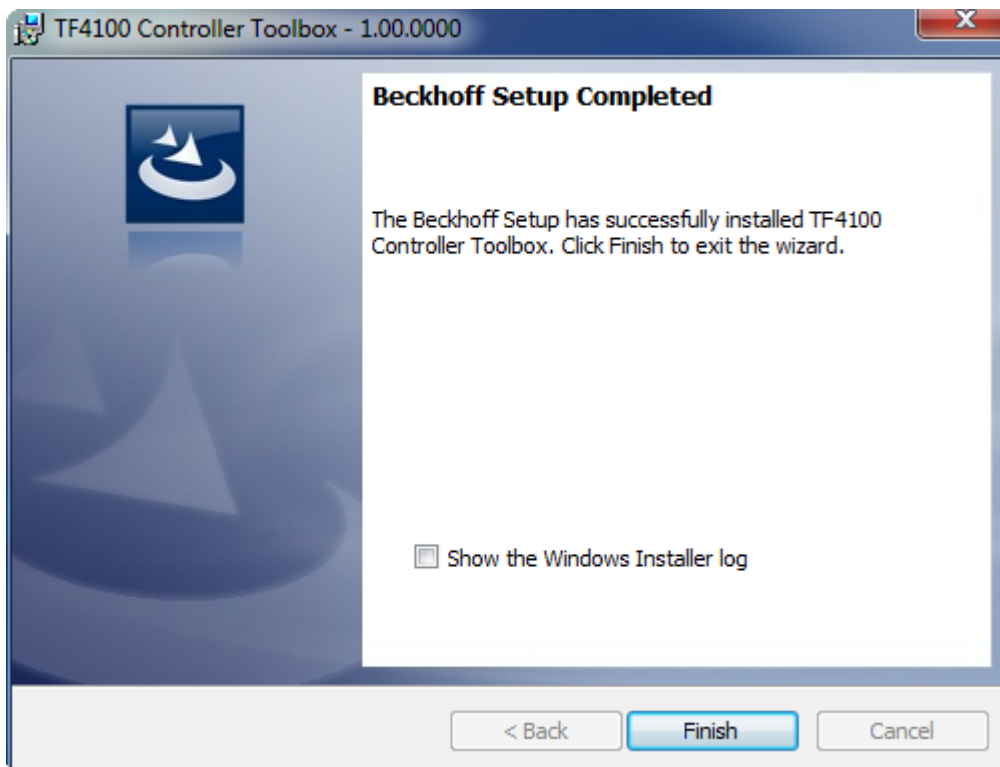5. Select **Next**, then **Install** to start the installation.

⇨ A dialog box informs you that the TwinCAT system must be stopped to proceed with the installation.

6. Confirm the dialog with **Yes**.

7. Select **Finish** to exit the setup.

⇨ The TwinCAT 3 Function has been successfully installed and can be licensed (see Licensing).

4 Technical introduction

4.1 Digital filters

Digital filters are used to manipulate digitalized (time-discrete and value-quantized) signals. The manipulation is evident in the frequency domain, where certain components of a signal are emphasized or suppressed.

Properties

Digital filters can differ, among other things, in the frequency domain that may pass through the filter.

Filter type	Description	Area of application (examples)
Low-pass	Frequencies below a cut-off frequency can pass through the filter.	Anti-aliasing filter or filter for smoothing a signal.
High-pass	Frequencies above a cut-off frequency can pass through the filter.	Elimination of an interfering DC component in the signal.
Band-pass	Frequencies within a certain frequency interval can pass through the filter.	Useful for amplitude-modulated signals (radio technology, optical measuring signals, ultrasound signals, ...), i.e. the wanted signal is spectrally distributed around a carrier frequency, so that low and high frequencies outside the wanted signal worsen the SNR (signal-to-noise ratio) and are suppressed.
Band-stop	Frequencies out of a certain frequency interval can pass through the filter.	Suppression of an inductively coupled frequency, e.g. the mains frequency.

The specific implementation of the filter determines the transition behavior from the passband to the stopband.

See also: Filter types and parameterization [▶ 14]

Digital signals

An analog signal x(t) is converted by an analog-to-digital converter, e.g. in an EL3xxx or ELM3xxx, to a time-discrete and value-quantized signal x[n]. The time discretization takes place with the sampling period T (inverse of the sampling rate f_s).

$$x[n] = x(t = nT)$$

Difference equation

The general difference equation for an input signal x[n] (input to a discrete system, in this case a filter) and a corresponding output signal y[n] is:

$$a_0 y[n] + \sum_{k=1}^{N} a_k y[n-k] = \sum_{k=0}^{M} b_k x[n-k]$$

a_k und b_k are usually real-valued coefficients (filter coefficients). The current output value y[n] of a system is thus calculated as a linear combination of past filter inputs x[n-k] with k > 0, past filter outputs y[n-k] with k > 0 and the current filter input x[n] (k = 0).

The inclusion of past filter outputs in the calculation of a current output value represents a feedback and therefore requires verification to ensure system stability. Filters with feedback are called "IIR filters" (Infinite Impulse Response filters). Filters without feedback are called "FIR filters" (Finite Impulse Response filters). The advantage of IIR filters is that "good" manipulations of the signal x[n] can be achieved with low filter orders. By definition, FIR filters can never be unstable.

Transfer function

By z-transforming the difference equation and using the linearity and the time shift property, the following general representation of the filter transfer function is obtained:

$$G(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{k=0}^{M} b_k z^{-k}}{\sum_{k=0}^{N} a_k z^{-k}} = \frac{b_0 + b_1 z^{-1} + b_2 z^{-2} + \dots + b_M z^{-M}}{a_0 + a_1 z^{-1} + a_2 z^{-2} + \dots + a_N z^{-N}}$$

The denominator coefficients a_k belong to the coefficients in the feedback. In order for the filter to be stable in conjunction with the transfer function G(z), care must be taken when calculating these coefficients that the poles of G(z) lie within the unit circle in the complex level.

IIR filters with a high filter order can become instable due to quantification effects during the calculation of the coefficients. In order to overcome these challenges, IIR filters are often implemented in cascaded biquad filters, normally called second-order sections (SOS). The overall transfer function is expressed by a multiplication of several 2nd order filters. The transfer function G(z) is then described with:

$$G(z) = \prod_{m=1}^{M} G_m(z) = \prod_{m=1}^{M} \frac{b_{0m} + b_{1m} z^{-1} + b_{2m} z^{-2}}{a_{0m} + a_{1m} z^{-1} + a_{2m} z^{-2}}$$

The frequency response of a system can be determined from the transfer function G(z) by transitioning to the frequency range (frequency f) with $z = \exp(j2\pi f T)$. The amplitude response then corresponds to the magnitude of the frequency response, and the phase response corresponds to the argument of the frequency response.

Implementation in the PLC library

The PLC library Tc3_Filter provides various function blocks for implementing digital filters.

The function block FB_FTR_IIRCoeff can be used to implement a free filter. The filter coefficients a_k and b_k can be calculated individually and transferred to the function block using a configuration structure. You are responsible for the stability of your filter.

With the function block FB_FTR_IIRSos you can implement a free filter structured in SOS. The filter coefficients a_k and b_k can be calculated individually and transferred to the function block using a configuration structure. You are responsible for the stability of your filter.

The function block FB_FTR_IIRSpec can be used to implement ready-made filters of type Butterworth or Chebyshev through simple parameterization. The filter coefficients are thereby calculated internally as biquads.

The function block FB_FTR_MovAvg can be used to implement a moving average filter, which is used in many applications for smoothing signals.

In addition, further filters that are commonly used in system theory and control technology are made available to you: PT1, PT2, PT3 and PTn elements.

A PT1 element and a Butterworth 1st order low-pass filter can be converted equivalently to each other, but the featured parameters of the filters are different.

Bilinear transformation

The parameterization of the predefined filters takes place in the Laplace space. The implementation of the time-continuous system representation in the time-discrete z-space takes place internally with the help of the bilinear transformation.

$$s = \frac{2}{T}\frac{z-1}{z+1}$$

The effect of "frequency warping" is taken into account in the filter design.

4.2   Filter types and parameterization

i This description is limited to low-pass filters. However, the concepts can be applied to other filter types (high-pass, band-pass and stop-band filters).

The Butterworth filter and the Chebyshev filter are common implementations of a digital filter.

The difference between the two implementations essentially consists of the balance between the permissible ripple of the amplitude response in the passband and the slope of the amplitude response in the transition between the passband and the stopband. While the Butterworth filter has a maximally flat amplitude response in the passband, for the Chebyshev filter the permissible ripple of the amplitude response in the passband is specified as a parameter. The advantage of the Chebyshev filter is a steeper decrease of the amplitude response in the transition range from the passband to the stopband.

The filter types are compared and described in more detail below. First, some basic terms are explained briefly.

Transfer function in the amplitude/frequency diagram

The filter is described mathematically by the transfer function (see <u>Digital filters [▶ 12]</u>). The transfer function can be displayed in the form of an amplitude and a phase response.
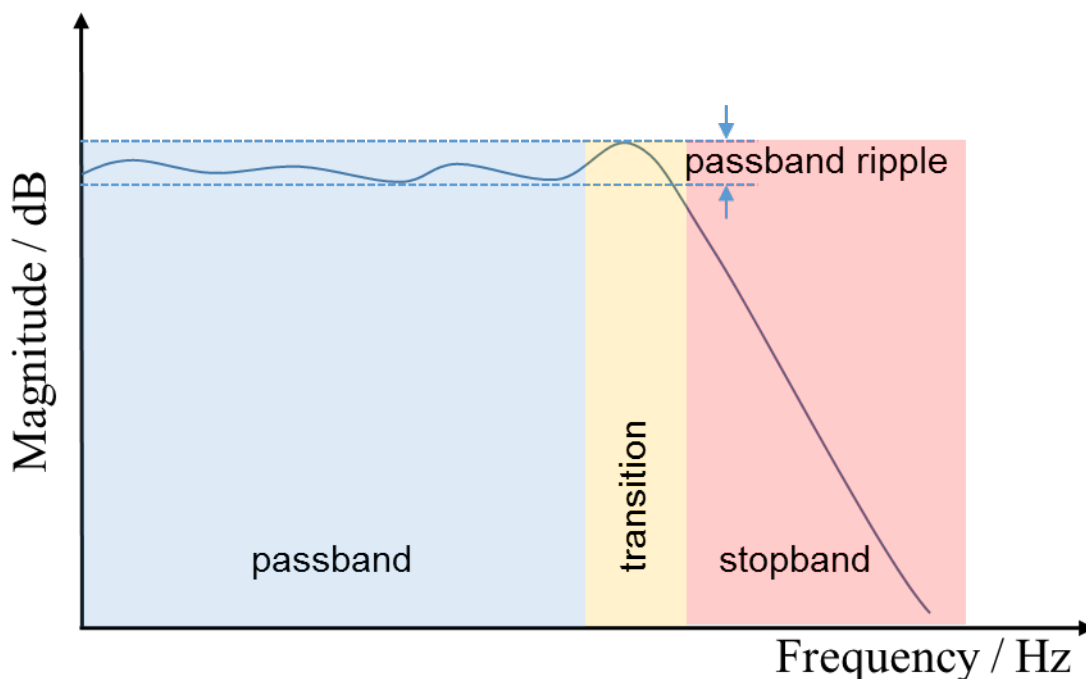

Fig. 1: Graphical representation of the amplitude response of a low-pass filter

Passband

The passband (blue zone) allows spectral components of a signal to pass through. Modification of the signal in this frequency range should be avoided.

Stopband

In the stopband (red zone), the filter attenuates the corresponding frequency components of the signal.

Transition

The transition (yellow zone) separates the passband and the stopband. It should normally be as small as possible. The design of the transition phase is a defining criterion for the selection of the filter type and its parameterization.

Passband ripple

The ripple in the passband describes the waviness of the amplitude response in the passband.

Parameterization of the Butterworth filter

Properties

The amplitude response of the Butterworth filter is maximally flat in the passband, so that the wanted signal in this range is only minimally manipulated. In addition, the entire course of the amplitude response is monotonous, i.e. without passband ripple. This filter type is therefore one of the most frequently used filter types.

Parameter

The transfer function of the Butterworth filter contains only two parameters that have to be defined: the cut-off frequency and the filter order.

Filter order

The filter order determines how steeply the amplitude response decreases in the transition range. The higher the filter order, the steeper the amplitude response decreases and the smaller the transition range.

The following applies for the slope of the amplitude response of a Butterworth filter: -n x 20 dB/decade, with n = order, i.e. -20 dB/decade for filter order 1, -40 dB/decade for filter order 2, etc.
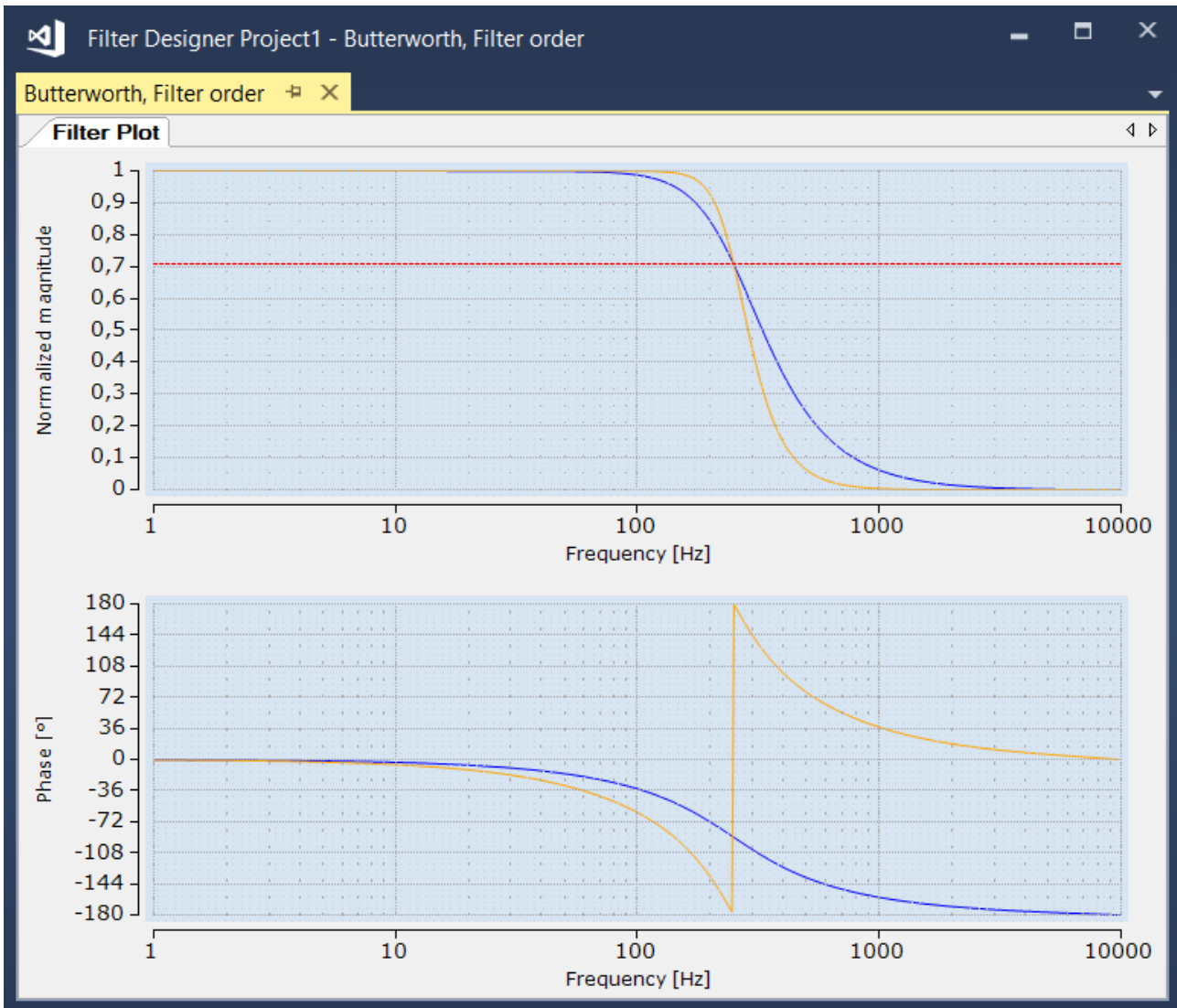
Fig. 2: Graphical representation of the amplitude and phase response of a Butterworth filter (blue: filter order 2, yellow: filter order 4)

Cut-off frequency

The cut-off frequency of the Butterworth filter is defined according to its transfer function as the frequency at which the normalized amplitude response assumes the value 1/sqrt(2) ≈ -3 dB. This applies to all filter orders. Accordingly, when designing the filter, care must be taken to ensure that the spectral components of a signal are already attenuated by 3 dB at the cut-off frequency. This parameter causes a parallel shift of the amplitude response along the frequency axis (distortion due to the logarithmic frequency axis).
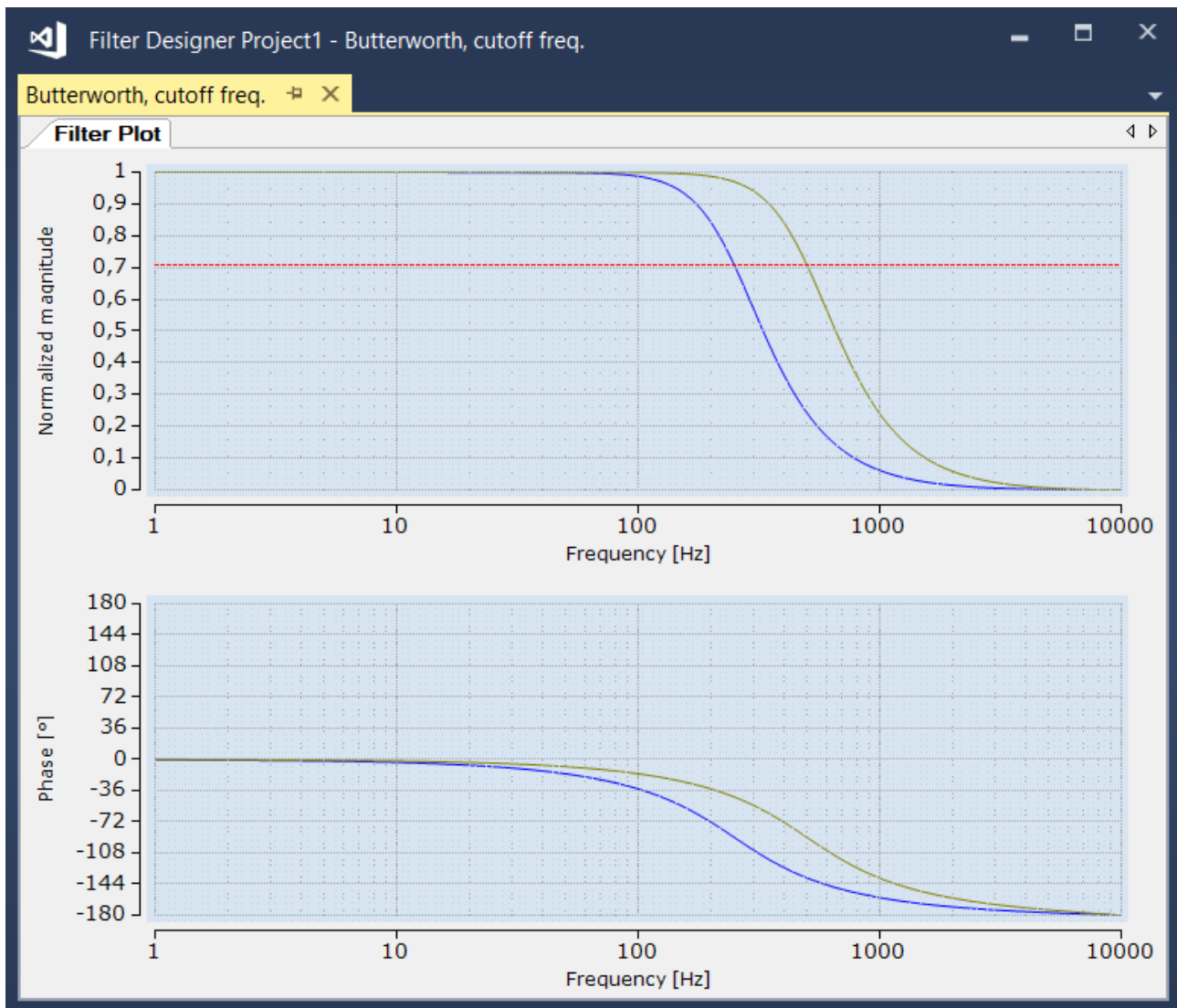
Fig. 3: Graphical representation of the amplitude and phase response of a Butterworth filter (blue: cut-off frequency 400 Hz, green: cut-off frequency 700 Hz)

Parameterization of the Chebyshev filter

Properties

The amplitude response of the Chebyshev filter has a parameterizable passband ripple. However, the amplitude response decreases steeply in the transition range when the filter order is small. The following applies: The greater the permissible passband ripple, the shorter the transition range.

Parameter

In addition to the filter order and the cut-off frequency as parameters to be defined, the transfer function of the Chebyshev filter contains a "passband ripple" parameter. This also affects the definition of the cut-off frequency.

Passband ripple

The parameter specifies the permissible ripple of the amplitude response in the passband of the filter. By allowing a passband ripple, a short transition range between passband and stopband, and thus a steep decrease of the amplitude response, can be achieved with a significantly lower filter order.

Cut-off frequency

The cut-off frequency of the Chebyshev filter is defined as the frequency at which the amplitude response passes downwards through the defined "passband ripple".

The position of the transition range on the frequency axis is thus associated not only with the cut-off frequency, but also with the settings for the filter order and ripple.

The following diagram shows three different Chebyshev filters with different filter order and ripple, but the same cut-off frequency.
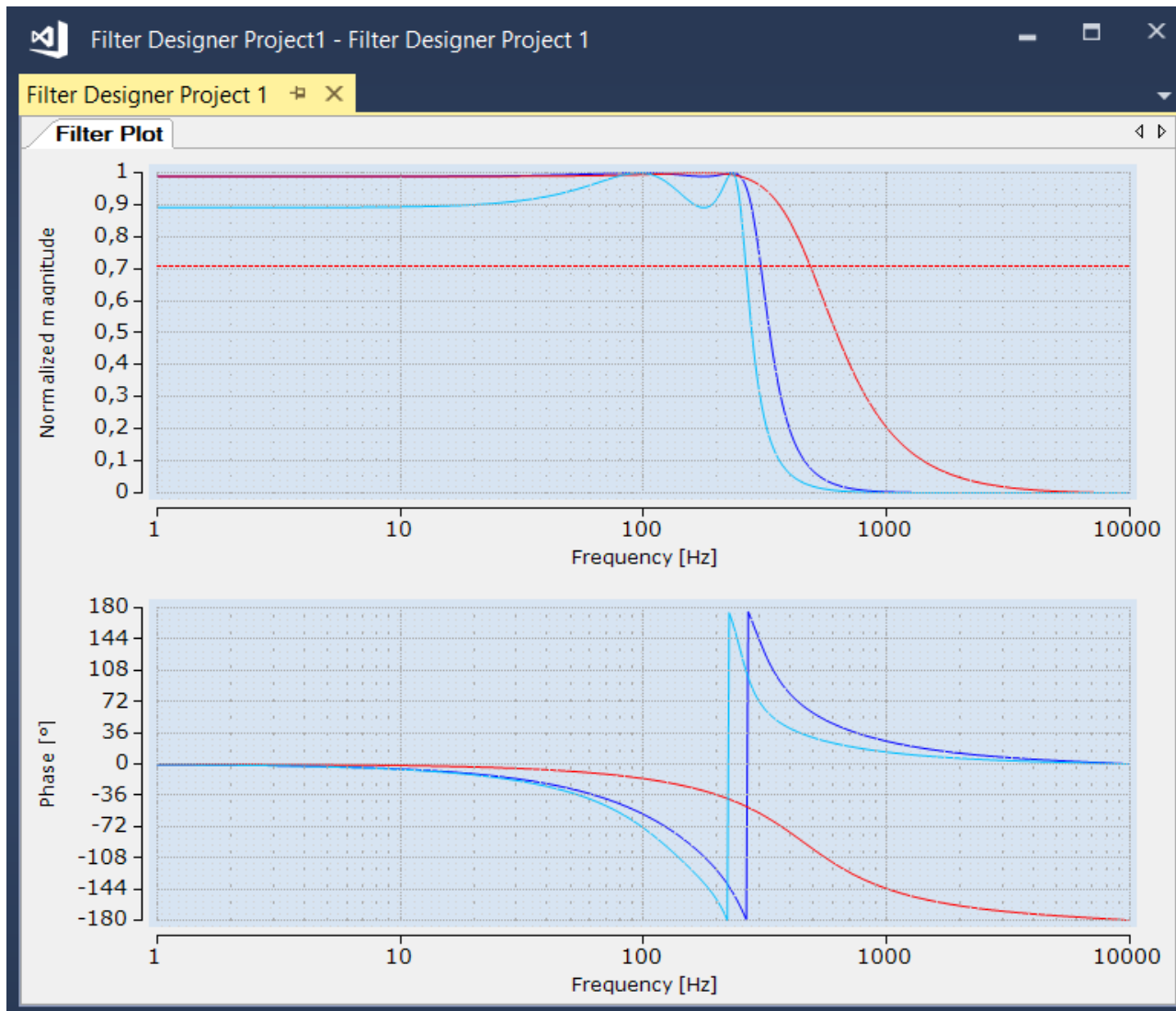

Fig. 4: Graphical representation of the amplitude and phase response of a Chebyshev filter (blue: filter order 4, passband ripple 0.1 dB, red: filter order 2, passband ripple 0.1 dB, cyan: filter order 4, passband ripple 1 dB)

Comparison of Butterworth and Chebyshev filters

The following diagram shows a direct comparison of the amplitude and phase response of a Butterworth filter and a Chebyshev filter. Both filters are parameterized so that their amplitude responses intersect at the cut-off frequency of the Butterworth filter at a normalized amplitude of 1/sqrt(2). Both filters are defined as fifth order filters. The passband ripple parameter of the Chebyshev filter is 0.5 dB.

The weighing up referred to above between the permissible passband ripple of the amplitude response and the slope in the transition with the same filter order becomes apparent. With the same filter order, the amplitude response of the Chebyshev filter decreases more sharply in the transition than that of the Butterworth filter. On the other hand, its amplitude response is not smooth in the passband, so that the wanted signal is manipulated more strongly here than with the Butterworth filter.
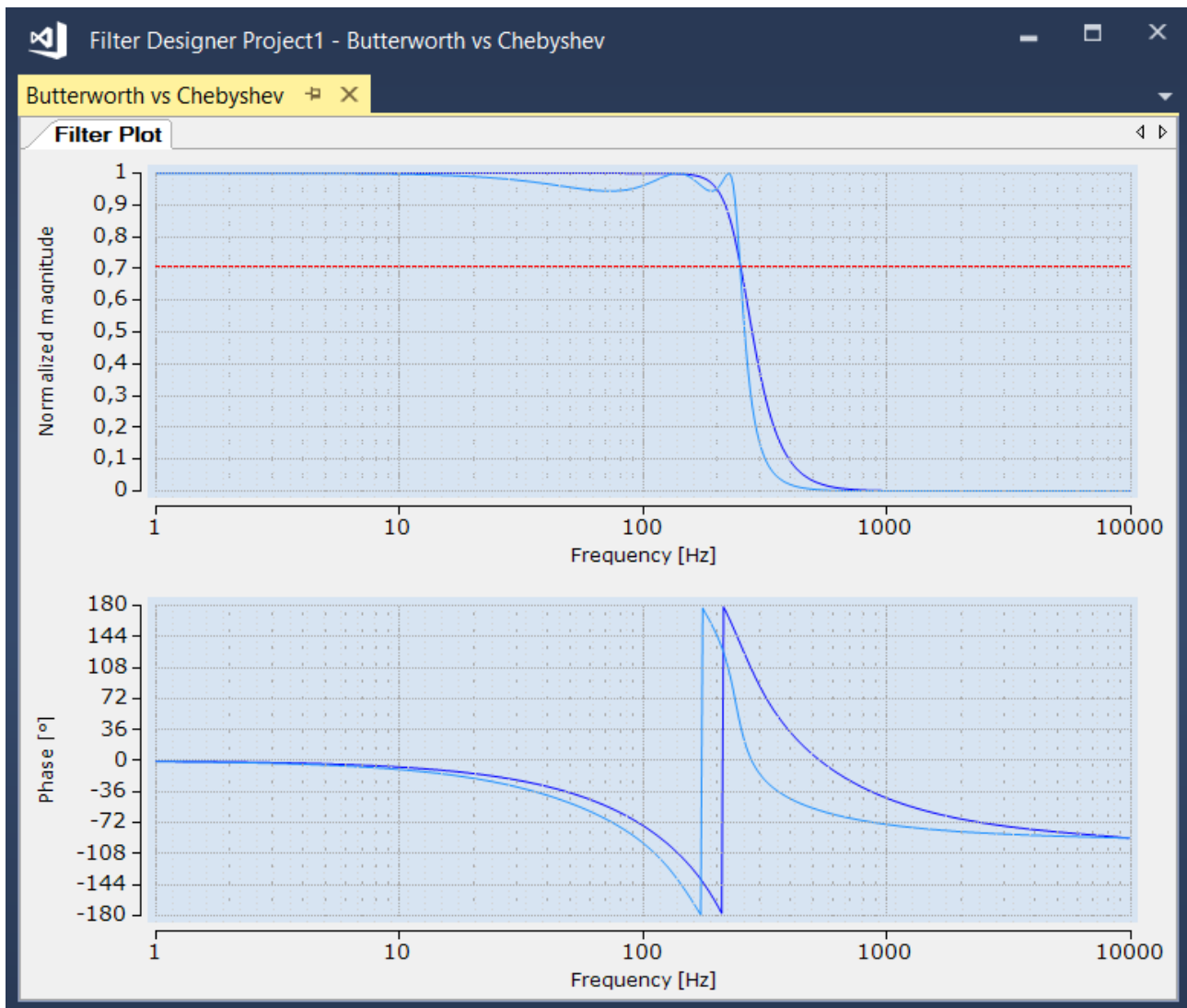
Fig. 5: Graphical representation of the amplitude and phase response of a Butterworth filter (blue) and a Chebyshev filter (cyan)

5 Configuration

5.1 New project

In order to be able to design digital filters, you must first create an appropriate project. There are various ways to do this. Fundamentally, you must first open the Visual Studio® **New Project** dialog.

In the **New Project** dialog you can directly select the **Filter Designer Project** and enter your project name (e.g. MyFilterProject). The project is then created in the Solution Explorer and, by clicking on an entry in the tree, the editor is displayed with the corresponding filter curve.
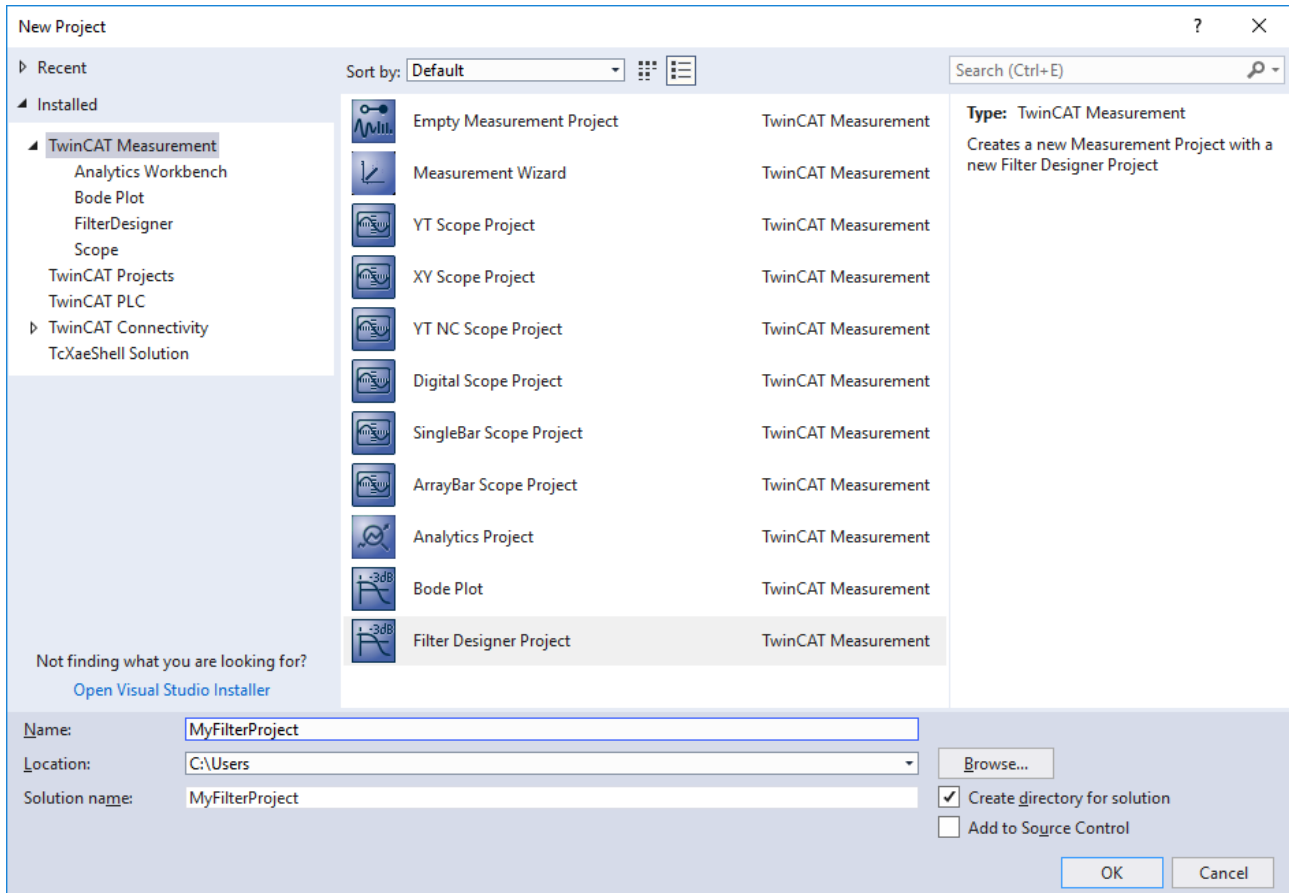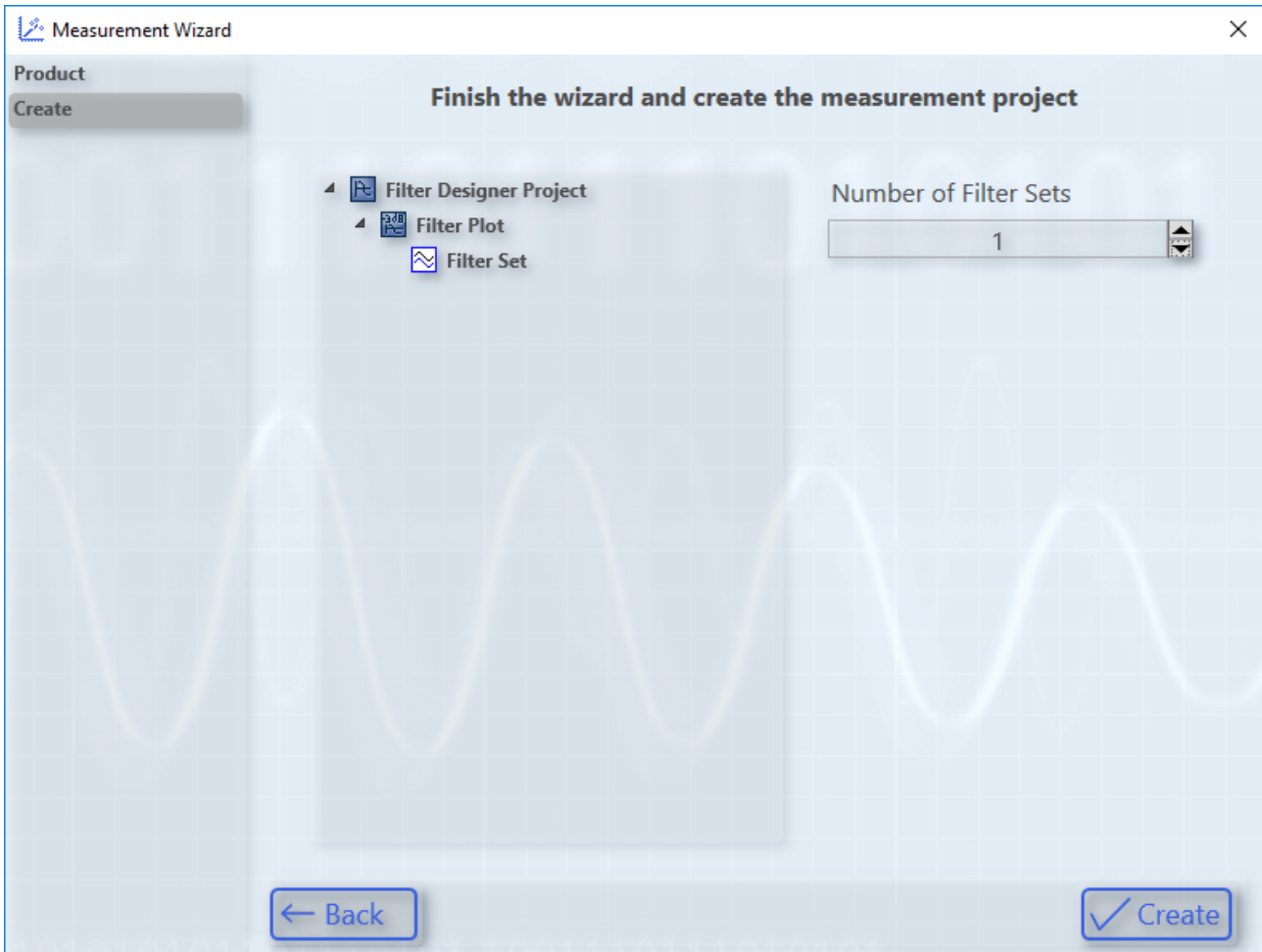

Alternatively, the **Measurement Wizard** can be selected.
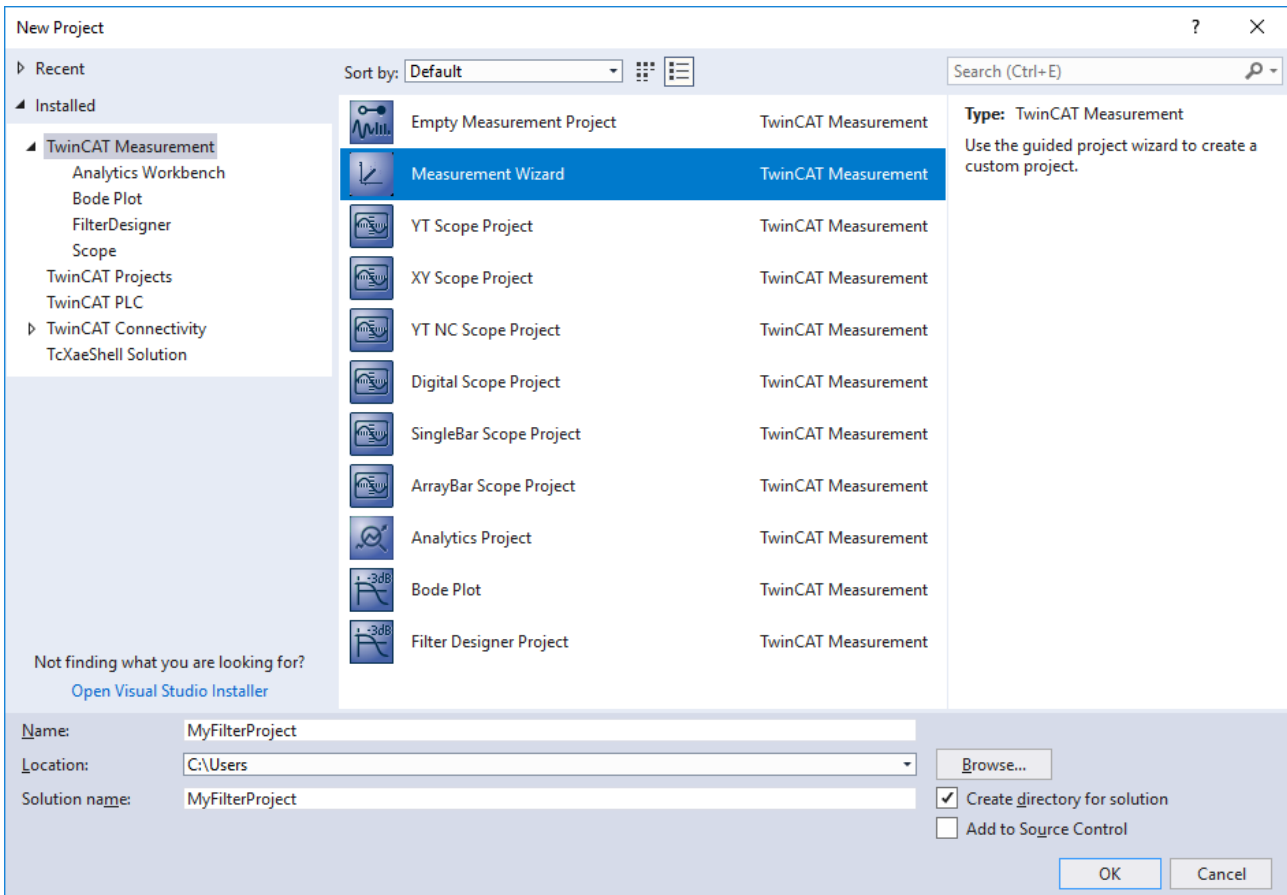
Among other things, it offers an entry for the **Filter Designer**,

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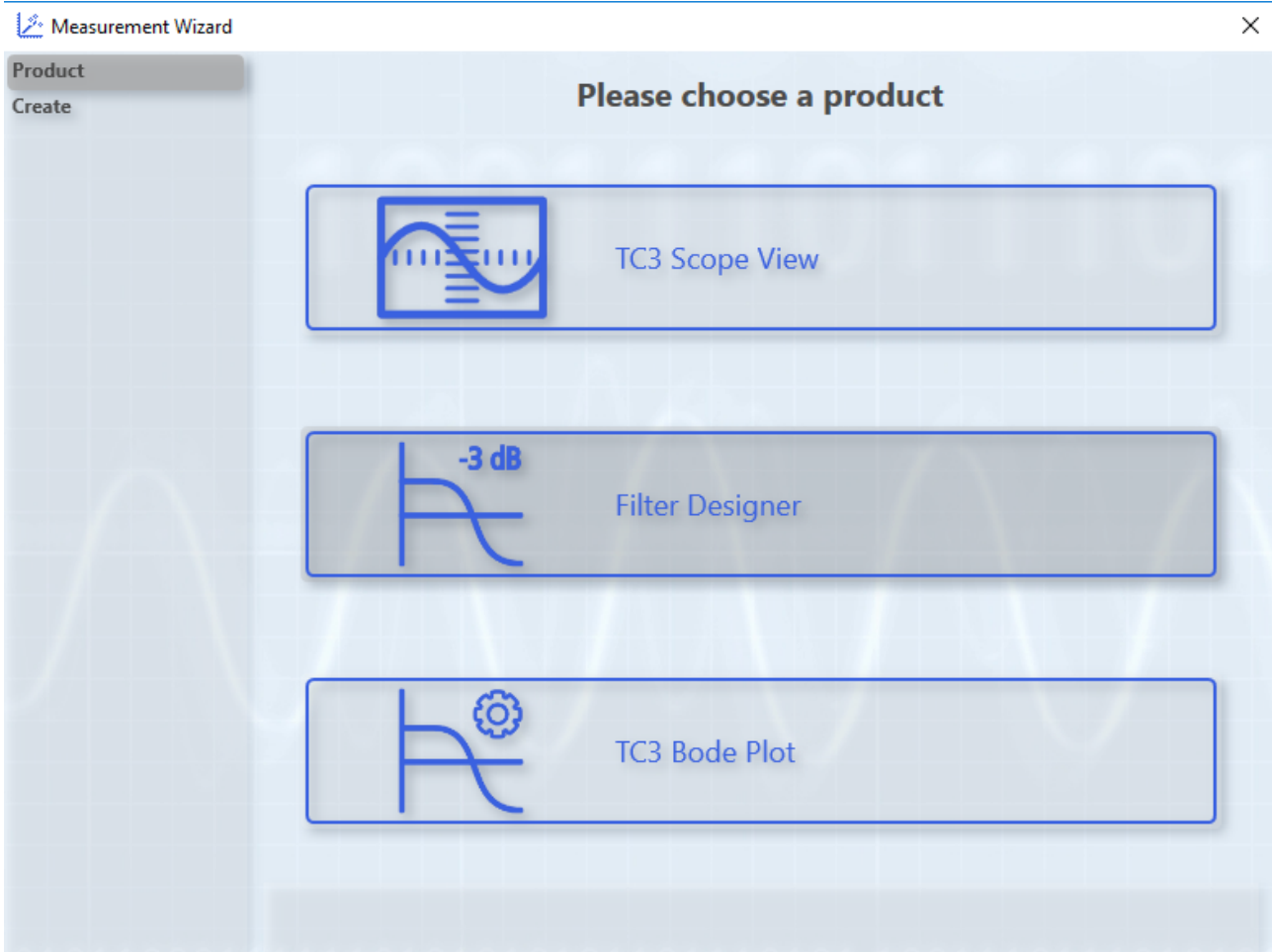
where you can specify how many filter curves are to be generated before creating the project.
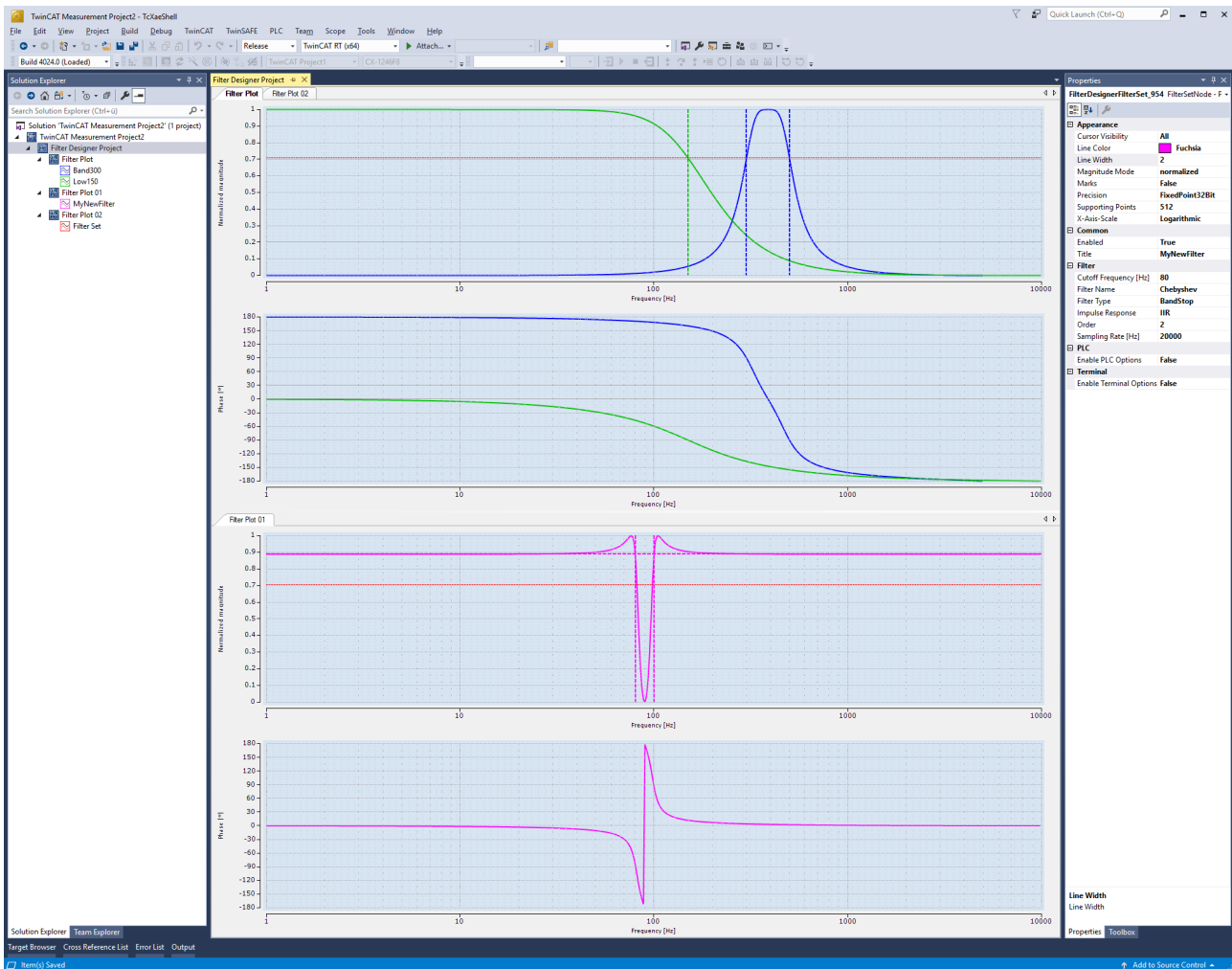

5.2   Filter

Depending on how you have created the filter project, at least one or more filter sets are visible in the Solution Explorer tree. A filter set maps a filter curve in the plot. You can add further filter sets to the plot at any time using the context menu.
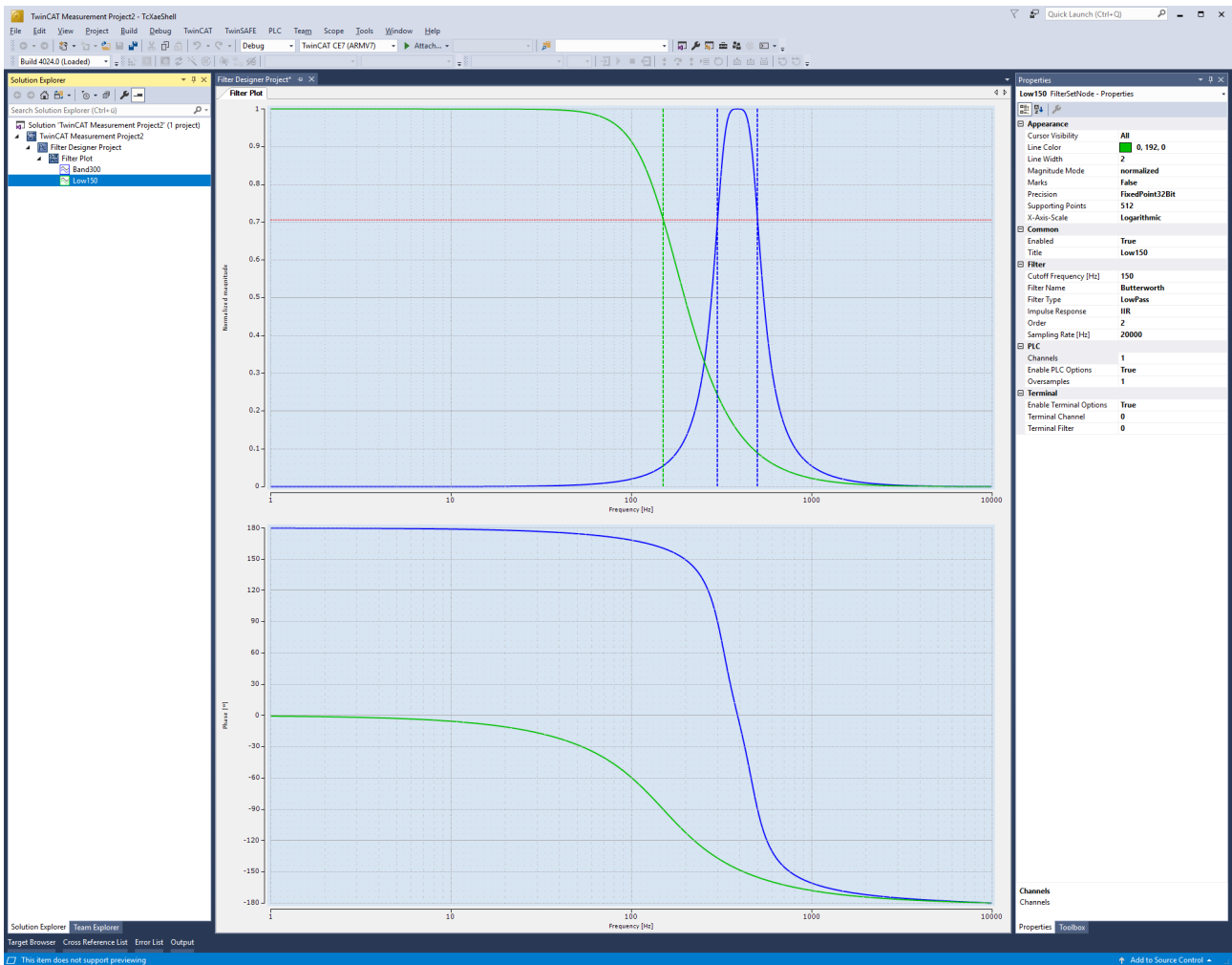
Of course, it is also possible to maintain several plots in parallel in a project.

The corresponding properties are described in the following sub-chapters, starting with the filter and the filter type.

5.2.1 Filter set properties

Some of the properties of the filter sets are independent of the filter and the filter type. These generally applicable filter set properties are described in more detail here.

Properties	▼ □ ✕
Band300 FilterSetNode - Properties	▾

⊟ **Appearance**

Cursor Visibility	**All**
Line Color	⬛ **Blue**
Line Width	**2**
Magnitude Mode	**normalized**
Marks	**False**
Precision	**FixedPoint32Bit**
Supporting Points	**1024**
X-Axis-Scale	**Logarithmic**

⊟ **Common**

Enabled	**True**
Title	**Band300**

⊞ **Filter**

⊟ **PLC**

Channels	**1**
Enable PLC Options	**True**
Oversamples	**1**

⊟ **Terminal**

Enable Terminal Options	**True**
Terminal Channel	**0**
Terminal Filter	**0**

Appearance

Properties | Toolbox

Appearance:

Cursor Visibility:	Specifies whether the drag & drop cursor for the cut-off frequency is displayed.
Line Color:	Specifies the color of the filter set.
Line Width:	Line width of the chart. Line width 1 causes the lowest computing time.
Magnitude Mode:	You can choose between two modes here. The default mode is normalized and corresponds to the magnitude of the frequency response. In dB mode, the frequency response is displayed with logarithmic scaling.
Marks:	Indicates whether the interpolation points of the characteristic curve should be displayed.
Precision:	The filter coefficients can be calculated with different precision. FixedPoint32Bit is the default mode and is required for the measuring terminals, since they calculate using fixed-point arithmetic. Alternatively, FixedPoint64Bit mode (higher precision) is to be selected if the coefficients are to be transferred to the PLC.
Supporting Points:	Number of interpolation points in the Bode Plot.
X-Axis-Scale:	The frequency axis can be scaled logarithmically or linearly.

Common:

Enabled:	With this property you can activate and deactivate the display of the filter curve.
Title:	Specifies the name of the filter set.

PLC:

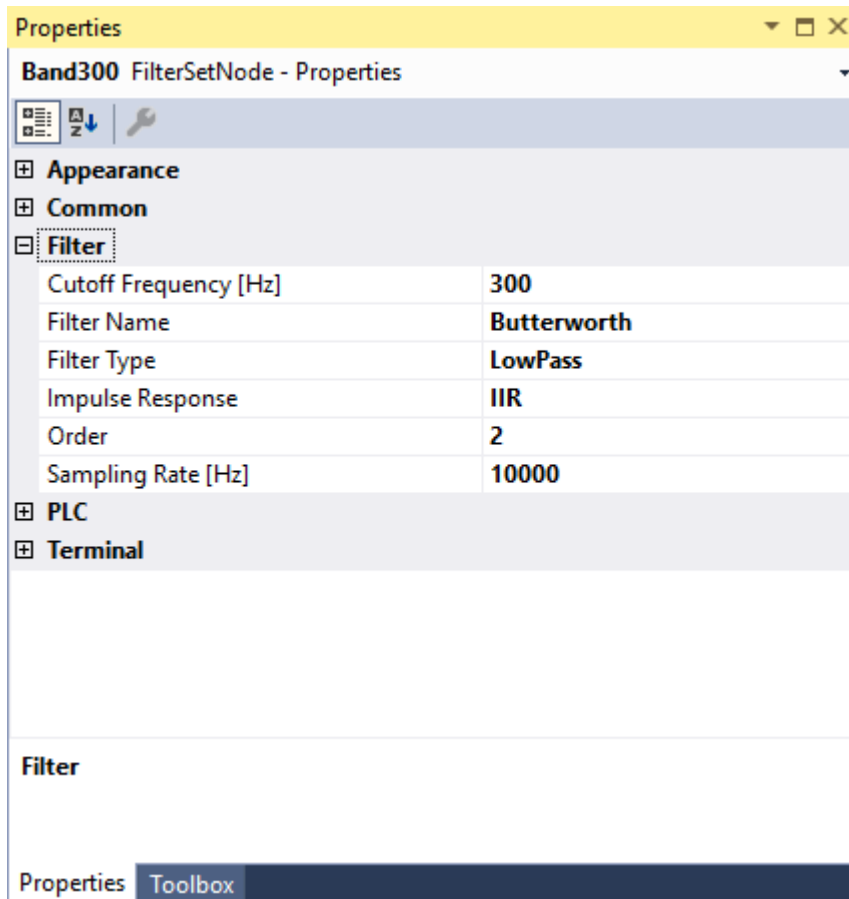| Channels:	Number of channels with which the filter structure is to be configured.
Enable PLC Options:	Must be true if a filter is to be transferred to the PLC.
Oversamples:	Number of oversamples with which the filter structure is to be configured.

Terminal:

Enable Terminal Options:	Must be true if a filter is to be transferred to the measuring terminal according to the channel and filter number.
Terminal Channel:	Specifies the channel number.
Terminal Filter:	Specifies the filter number.

5.2.1.1 Butterworth

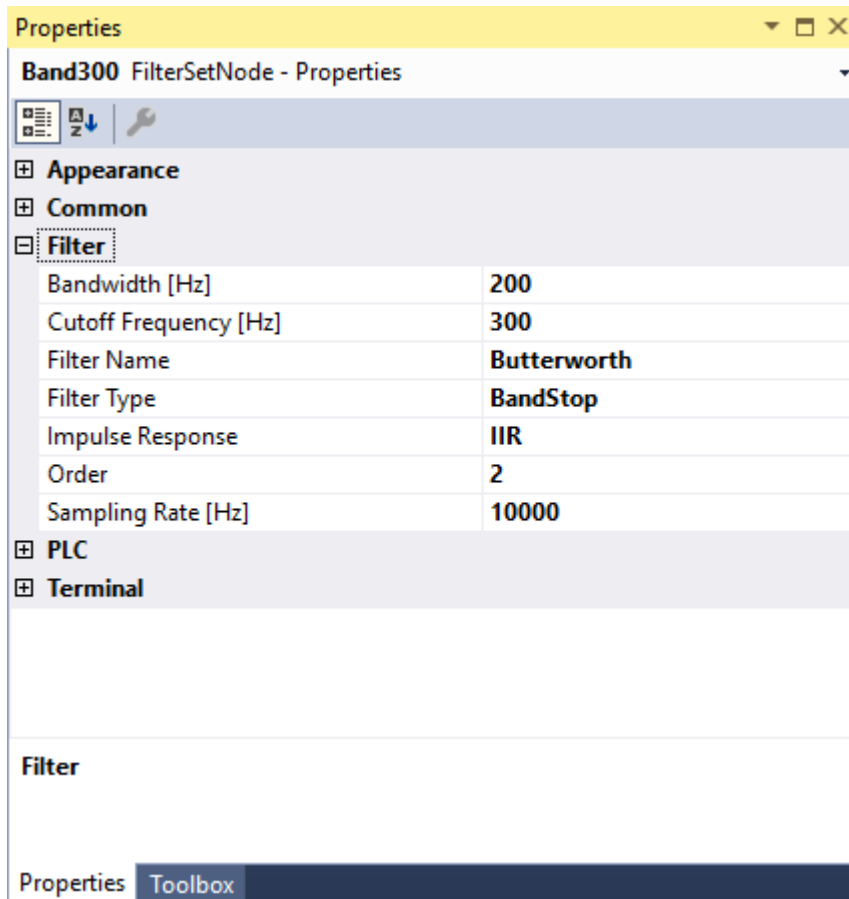The filter-specific properties of a Butterworth filter are described here.

Properties of a Butterworth low-pass and high-pass

Properties	▾ ▢ ✕
Band300 FilterSetNode - Properties	▾

▦ ▤↓ 🔧

⊞ **Appearance**
⊞ **Common**
⊟ **Filter**

Cutoff Frequency [Hz]	**300**
Filter Name	**Butterworth**
Filter Type	**LowPass**
Impulse Response	**IIR**
Order	**2**
Sampling Rate [Hz]	**10000**

⊞ **PLC**
⊞ **Terminal**

Filter

Properties   Toolbox

Filter

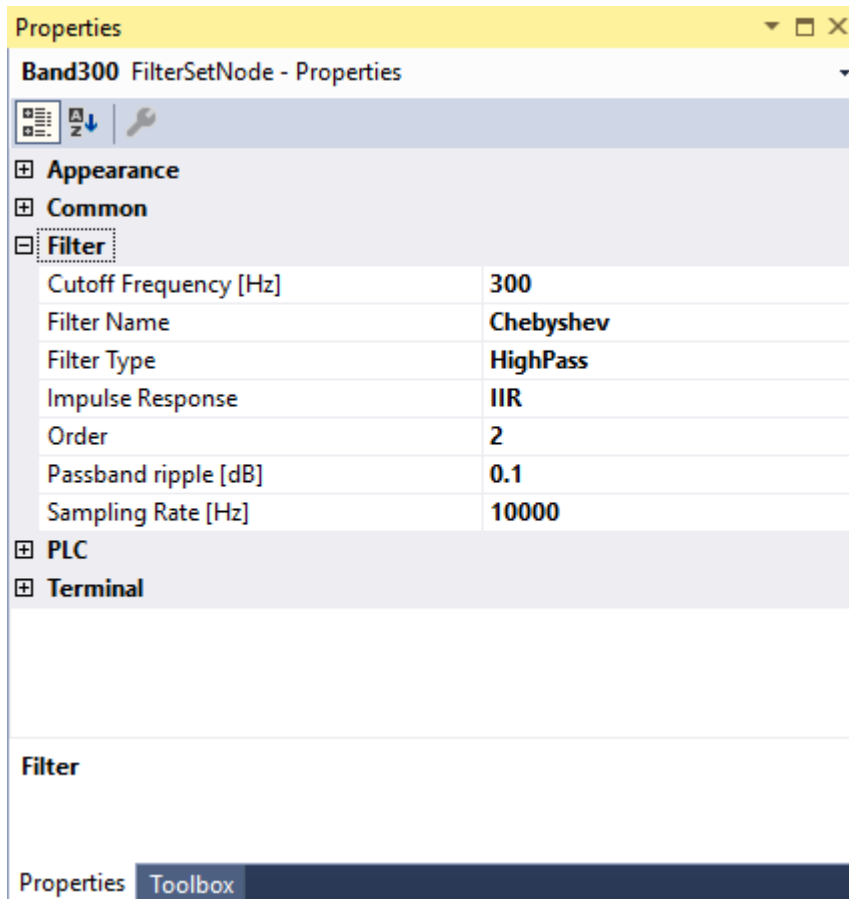| Cutoff Frequency [Hz]:	Specifies the cut-off frequency. This must be between 0 and the sampling rate/2.
Filter Name:	Butterworth should be selected here.
Filter Type:	Here you can select between low-pass, high-pass, band-pass and band-stop.
Impulse Response:	An IIR filter (Infinite Impulse Response Filter) can be chosen.
Order:	The order must be > 0 and ≤ 10 (band-pass/band-stop) or ≤ 20 (low-pass/high-pass).
Sampling Rate [Hz]:	The sampling rate must be greater than 0.

Properties of Butterworth band-pass and band-stop

Properties	▾ ☐ ✕
Band300 FilterSetNode - Properties	▾

⊞ **Appearance**
⊞ **Common**
⊟ **Filter**

Bandwidth [Hz]	**200**
Cutoff Frequency [Hz]	**300**
Filter Name	**Butterworth**
Filter Type	**BandStop**
Impulse Response	**IIR**
Order	**2**
Sampling Rate [Hz]	**10000**

⊞ **PLC**
⊞ **Terminal**

Filter

Properties | Toolbox

Filter

Supplementary properties for these types

Bandwidth [Hz]:	The bandwidth must be > 0 and < (sampling rate/2 – cut-off frequency).

5.2.1.2   Chebyshev

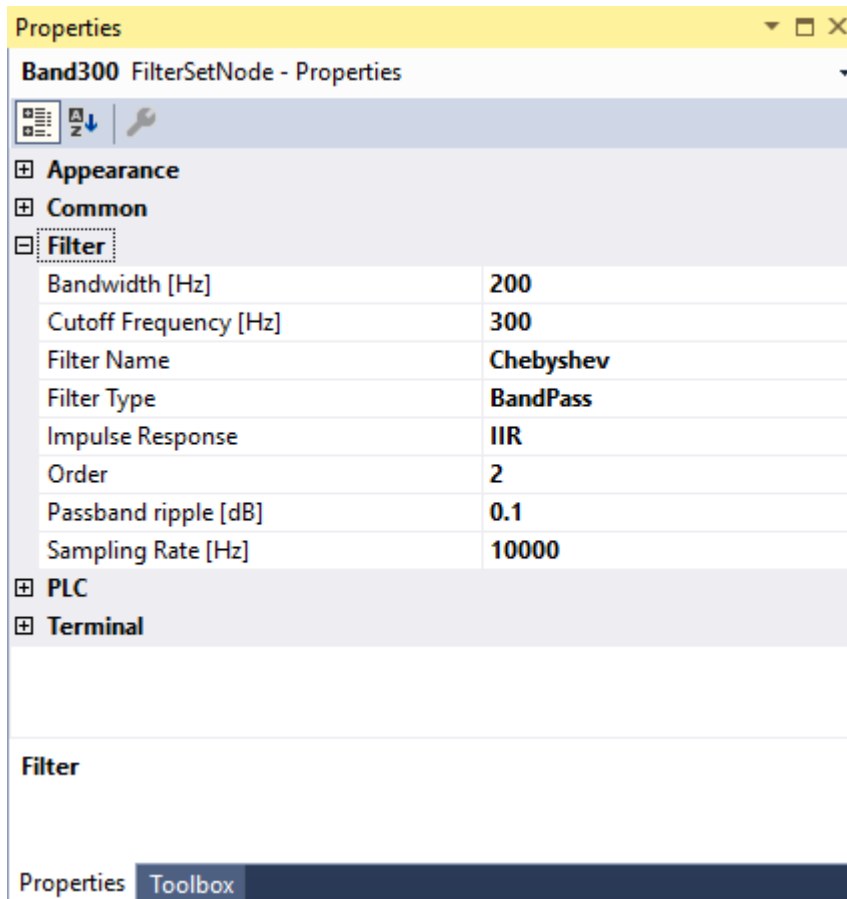The filter-specific properties of a Chebyshev filter are described here.

Properties of a Chebyshev low-pass and high-pass

Properties ▾ □ ✕

Band300 FilterSetNode - Properties ▾

⊞ **Appearance**

⊞ **Common**

⊟ **Filter**

Cutoff Frequency [Hz]	**300**
Filter Name	**Chebyshev**
Filter Type	**HighPass**
Impulse Response	**IIR**
Order	**2**
Passband ripple [dB]	**0.1**
Sampling Rate [Hz]	**10000**

⊞ **PLC**

⊞ **Terminal**

Filter

Properties | Toolbox

Filter

Cutoff Frequency [Hz]:	Specifies the cut-off frequency. This must be between 0 and the sampling rate/2.
Filter Name:	Chebyshev should be selected here.
Filter Type:	Here you can select between low-pass, high-pass, band-pass and band-stop.
Impulse Response:	An IIR filter (Infinite Impulse Response Filter) can be chosen.
Order:	The order must be > 0 and ≤ 10 (band-pass/band-stop) or ≤ 20 (low-pass/high-pass).
Passband Ripple [dB]:	The passband ripple (> 0) affects the ripple in the passband.
Sampling Rate [Hz]:	The sampling rate must be greater than 0.

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Properties of Chebyshev band-pass and band-stop

Properties	▾ □ ✕
Band300 FilterSetNode - Properties	▾

⊞ **Appearance**
⊞ **Common**
⊟ **Filter**

Bandwidth [Hz]	**200**
Cutoff Frequency [Hz]	**300**
Filter Name	**Chebyshev**
Filter Type	**BandPass**
Impulse Response	**IIR**
Order	**2**
Passband ripple [dB]	**0.1**
Sampling Rate [Hz]	**10000**

⊞ **PLC**
⊞ **Terminal**

Filter

Properties | Toolbox

Filter

| Bandwidth [Hz]:	The bandwidth must be > 0 and < (sampling rate/2 – cut-off frequency).

5.3   Use

The designed filter can be used in a variety of ways in TwinCAT. Firstly in interaction with the corresponding Filter PLC library and secondly with the measurement modules from the ELM series. Both ways are described in the following sub-chapters.

The application options can be viewed by right-clicking on the corresponding **Filter Set**, which opens the context menu.
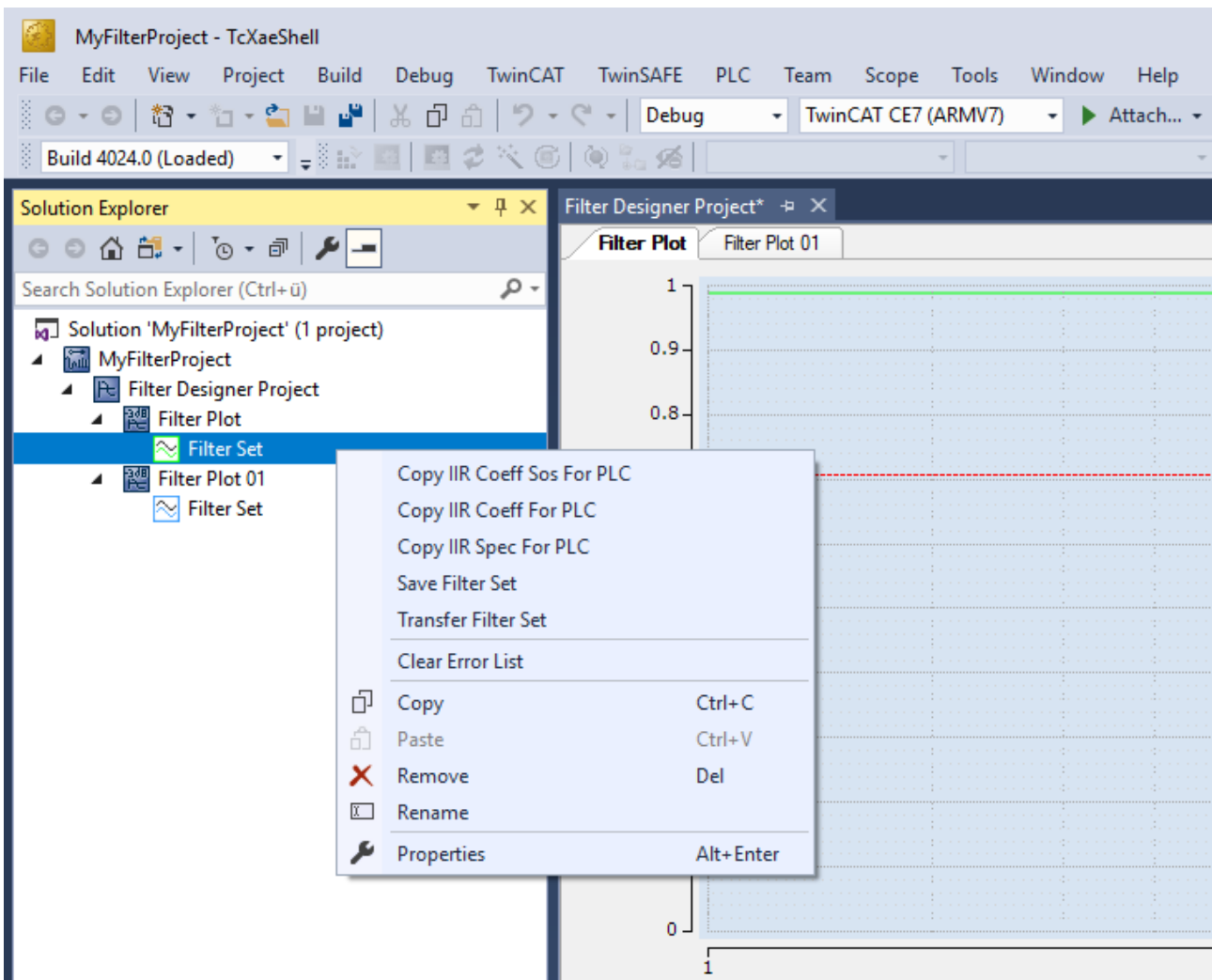
5.3.1 PLC

There are several ways of transferring the filters to the PLC in the form of specifications or coefficients. The simplest way is via the context menu, which is opened by right-clicking on the respective **Filter Set**.
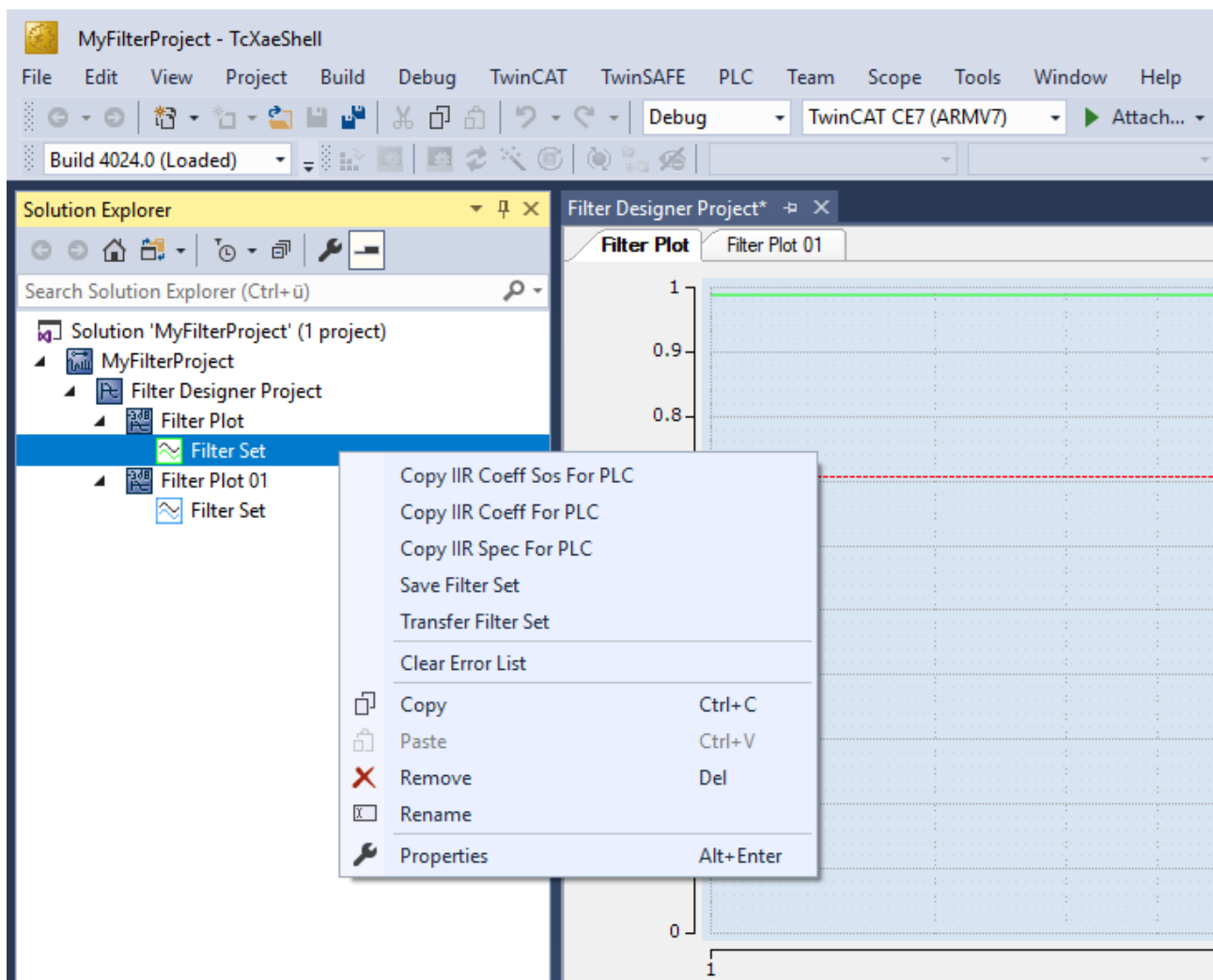
Copy the data of the designed filter to the clipboard, then go directly to the already existing PLC project, where you can paste the coefficients or corresponding specification. There are currently three different copy options.

Copy IIR Coeff Sos For PLC

Here, the coefficients are broken up into individual biquads and transferred for the FB_FTR_IIRSos in an array that must be specified in a structure, ST_FTR_IIRSos. The instance names of the array and the structure must be manually declared.

```
//CoeffArray_Sos

: ARRAY [1..1] OF ARRAY [1..6] OF LREAL := [[0.894858606122573, -1.78971721224515, 0.894858606122573
, 1, -1.77863177782458, 0.800802646665708]];

: ST_FTR_IIRSos :=(    nOversamples := 1,
                       nChannels := 1,
                       pCoefficientArrayAdr_Sos := ADR(),
                       nCoefficientArraySize_Sos := SIZEOF());
```

Copy IIR Coeff For PLC

Here, the coefficients are not broken up and are transferred for the FB_FTR_IIRCoeff in two arrays for the coefficients A and B. Instance names must be specified for both arrays. They are to be announced in the structure ST_FTR_IIRCoeff. The structure still requires an instance name.

```
//DenomCoeffArray_a
: ARRAY [1..3] OF LREAL := [1, -1.77863177782458, 0.800802646665708];
//DenomCoeffArray_b
: ARRAY [1..3] OF LREAL := [0.894858606122573, -1.78971721224515, 0.894858606122573];

: ST_FTR_IIRCoeff := (nOversamples := 1,
                      nChannels := 1,
                      pCoefficientArrayAdr_A := ADR(),
```

```
            nCoefficientArraySize_A := SIZEOF(),
            pCoefficientArrayAdr_B := ADR(),
            nCoefficientArraySize_B := SIZEOF());
```

Copy IIR Spec For PLC

Here, the set specification of the filter is copied in plain text in the form of the structure type ST_FTR_IIRSpec. The structure still requires an instance name.

```
: ST_FTR_IIRSpec :=(eFilterName := E_FTR_Name.eButterworth,
                eFilterType := E_FTR_Type.eHighPass,
                nFilterOrder := 2,
                fCutoff := 250,
                fSamplingRate := 10000,
                nOversamples := 1,
                nChannels := 1 );
```

A further option to transfer the filter properties to the PLC is offered by the **Transfer Filter Set** dialog.
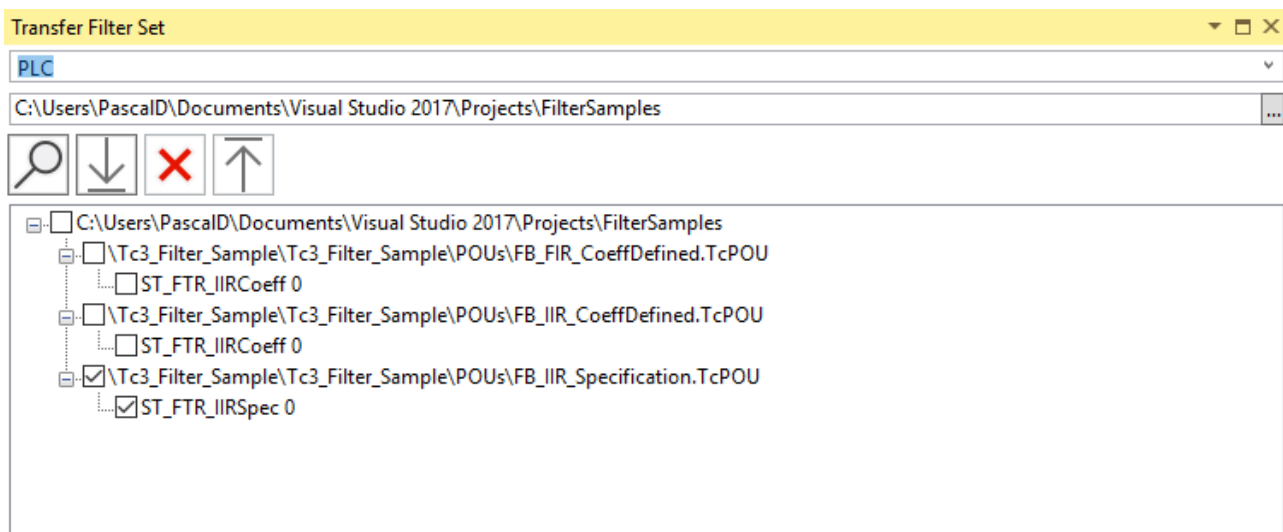

You can choose between PLC and I/O here. In the case of the PLC it is possible to specify a path, for example to a solution. This is checked for possible POUs of the Filter library, i.e. for instances of the Coeff or Spec structures. The scan procedure is accordingly initiated with the magnifying glass button.

Following the scan, all hits are displayed underneath in a tree view. You can now select which filter structure instance is to be overwritten by the designed filter. This function is also available if the scanned solution is already opened. Visual Studio® draws your attention to the fact that the file has changed externally and offers a reload.

5.3.2   I/O

There are two ways to transfer the filters in the form of coefficients to one or more terminals/channels (multi-download). You can transfer the filter by drag & drop or execute the **Transfer Filter Set** command via the context menu.
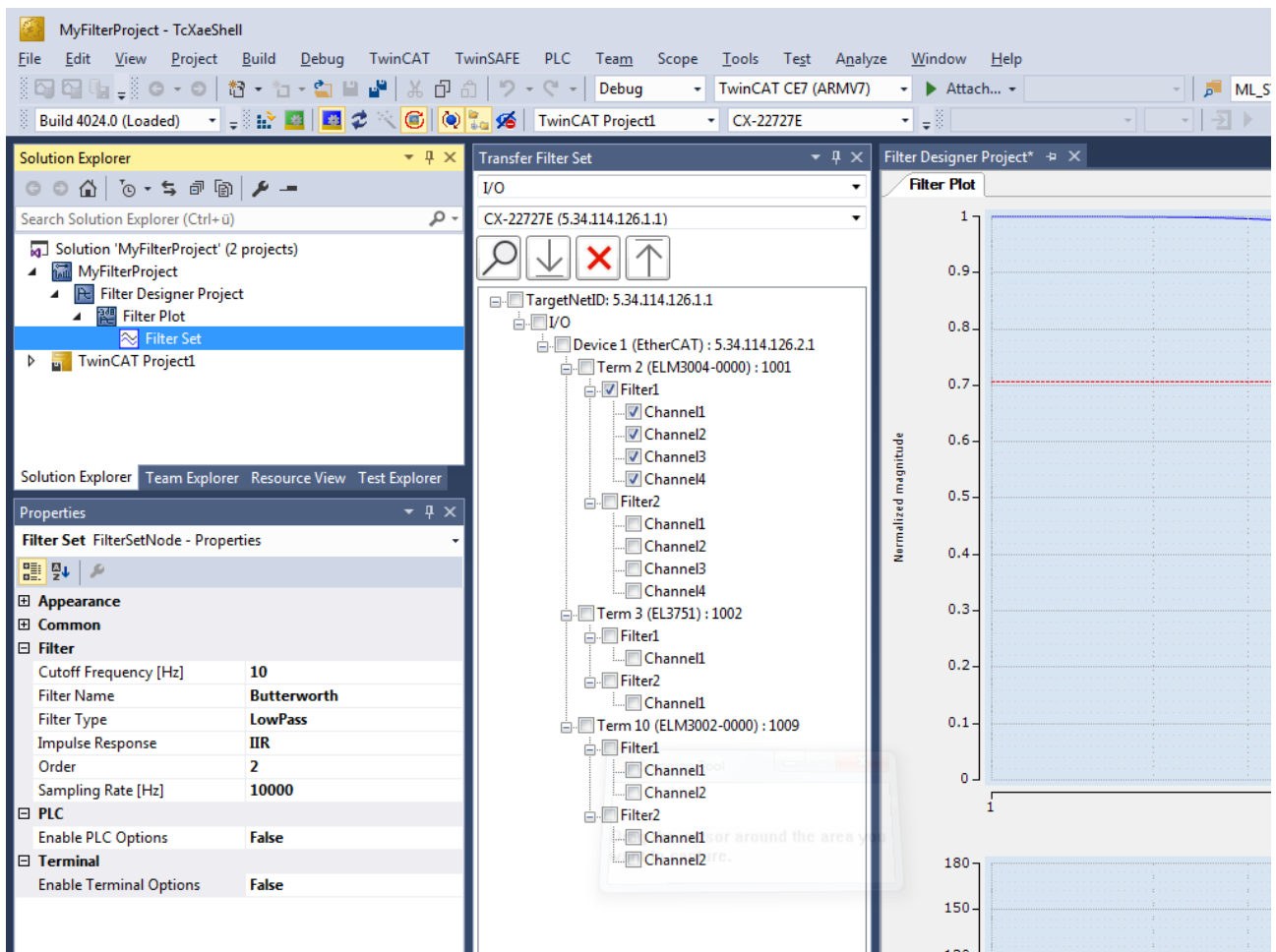
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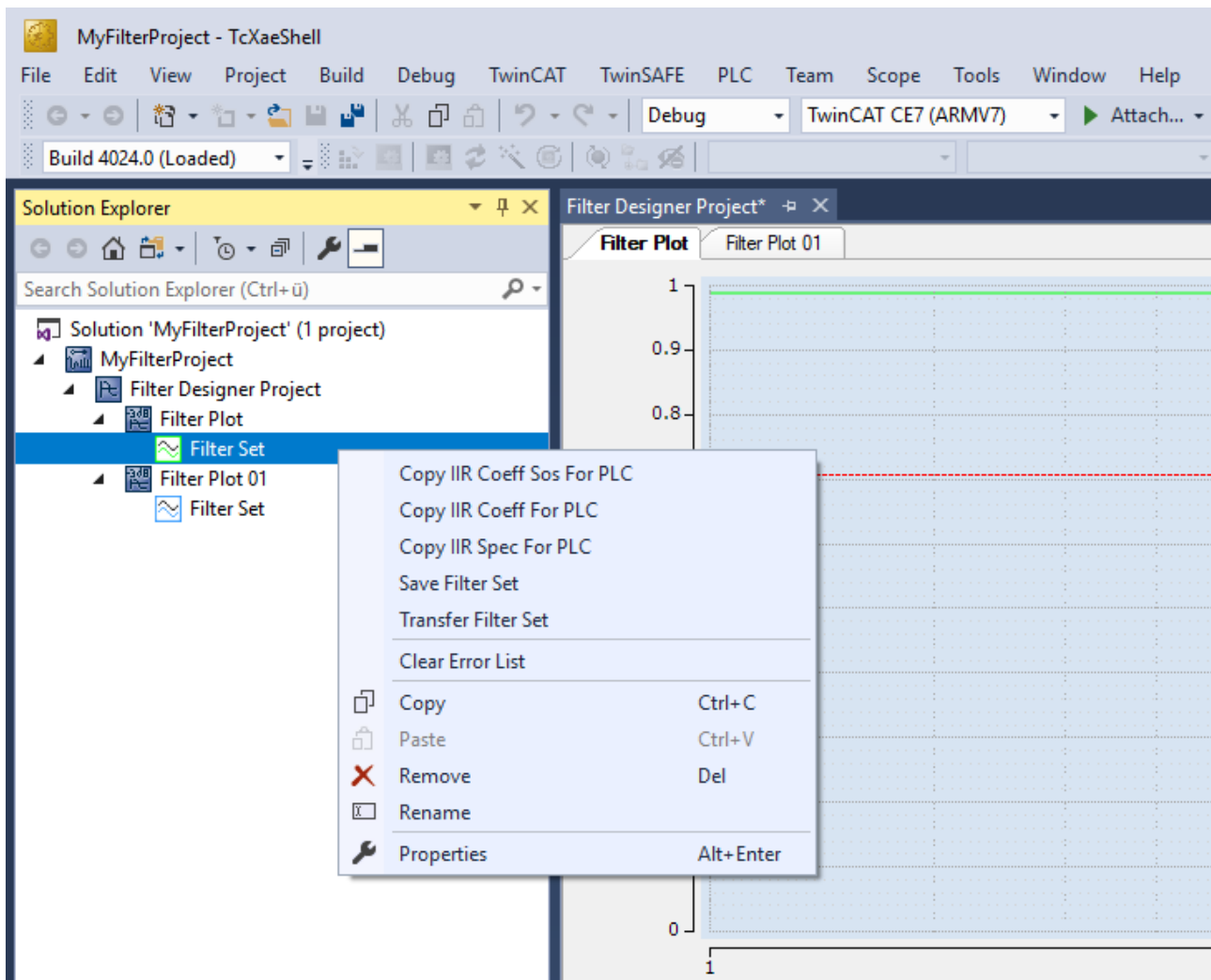
In the simplest case, the filter coefficients are transferred to the measuring terminal by dragging the created filter onto the desired channel by drag & drop.
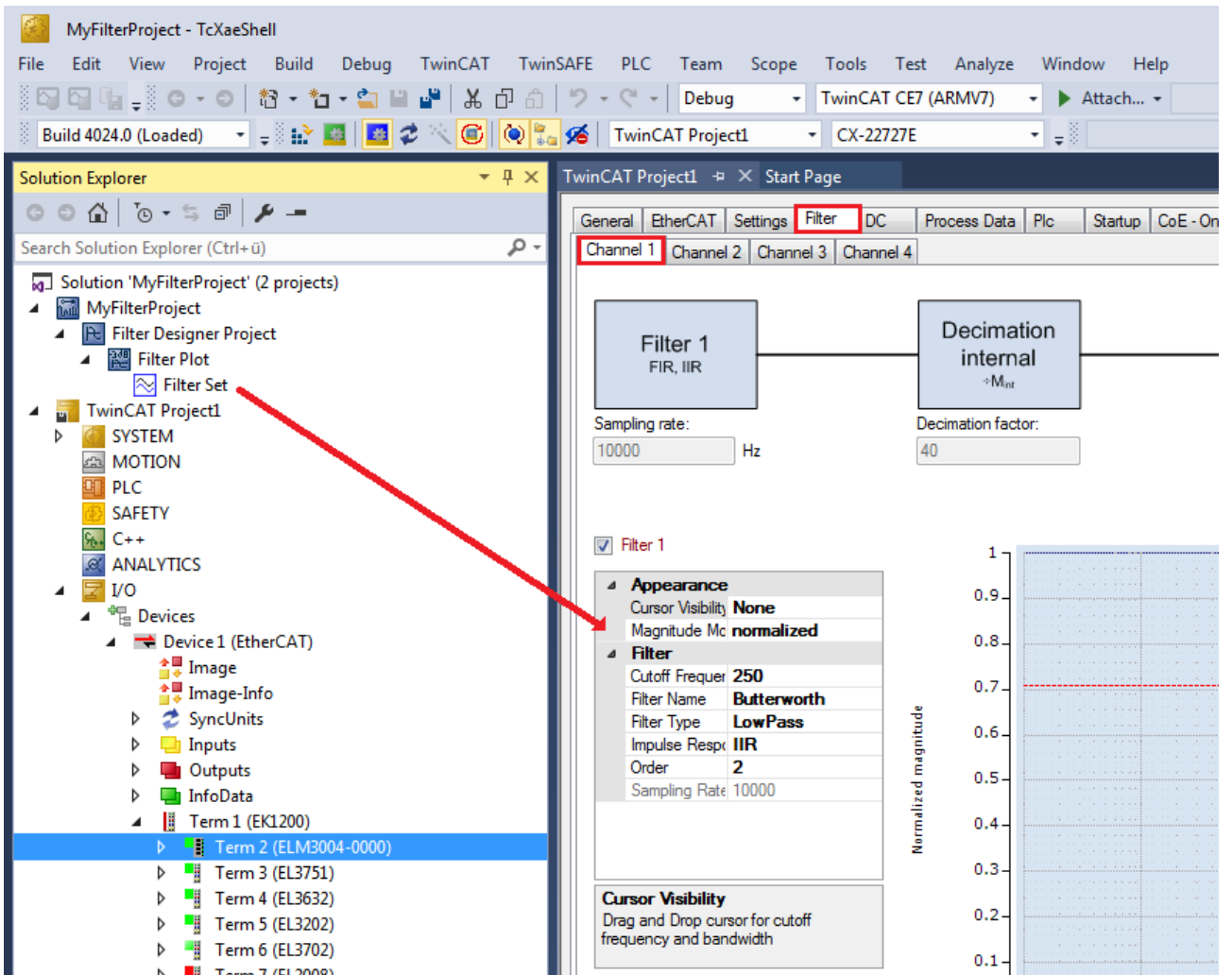
**Transfer of a single filter set (Enable Terminal Options = false)**

Alternatively, a filter can be transferred via the **Transfer Filter Set** dialog.

To do this, I/O and the target system must be selected with the appropriate NetId. Following the scan procedure, all existing measurement modules are displayed in a tree view. You can now select one or more channels and transfer the designed filter. In addition, the filters set in the terminal can be removed and also loaded to the Filter Designer.

Transfer of several filter sets (Enable Terminal Options = true)

It is possible to transfer several filter sets to different filter channels with the filter set property Enable Terminal Options.
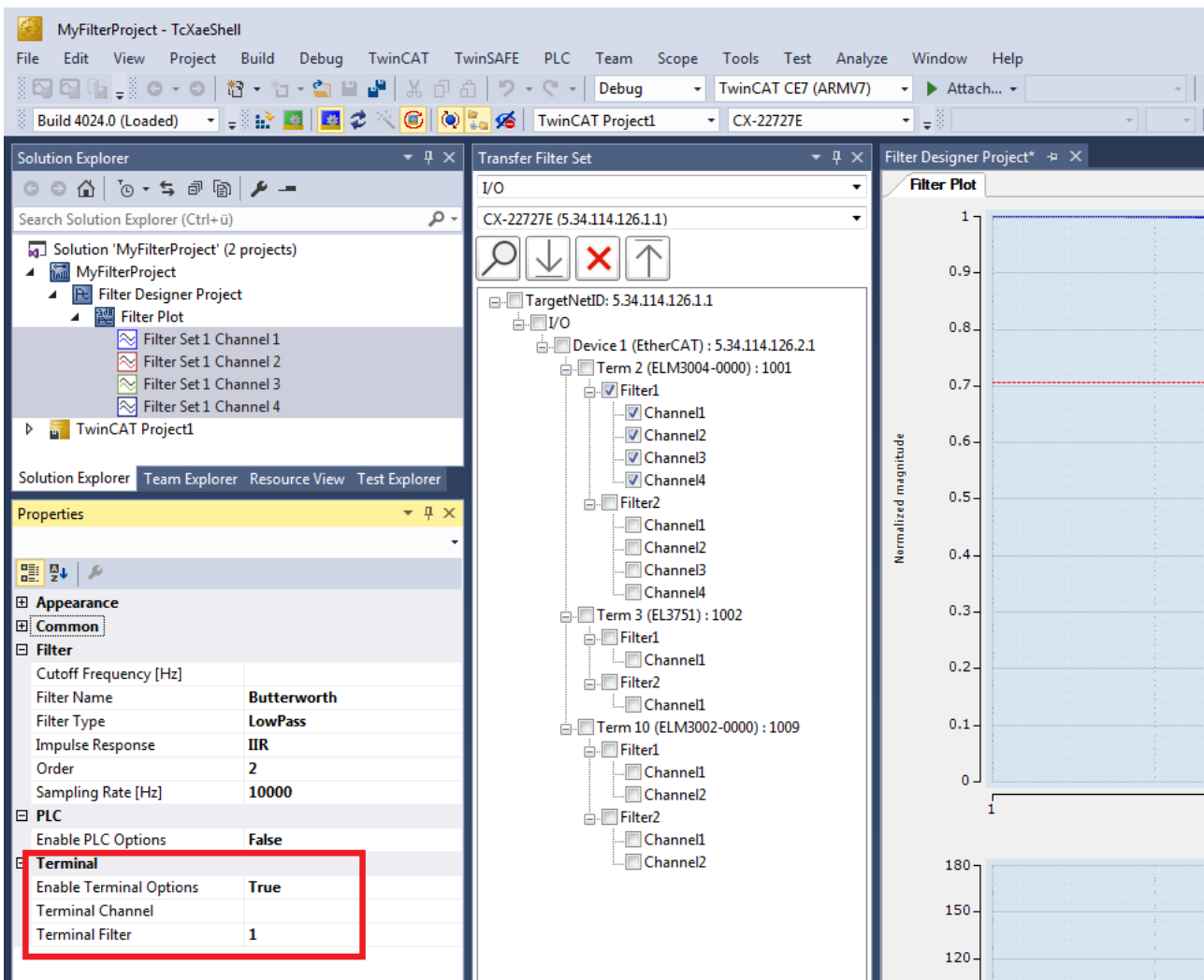
In this sample, four filter sets are marked with Terminal Filter = 1 and Terminal Channel from 1-4. Alternatively, the selection of the filter plot has the same effect, because all sub-members are then selected. In the Transfer Filter Set dialog, Channel1-Channel4 are selected under Filter1 for the ELM3004 terminal. A filter transfer leads to a transfer only taking place if the filter and channel numbers of the source (Filter Designer filter sets) and target (ELM terminal channel) correspond. This corresponds to the following four transfers:

- Filter Set 1 Channel 1 -> ELM3004:Filter1:Channel1
- Filter Set 1 Channel 2 -> ELM3004:Filter1:Channel2
- Filter Set 1 Channel 3 -> ELM3004:Filter1:Channel3
- Filter Set 1 Channel 4 -> ELM3004:Filter1:Channel4

Comment:

- The selected filter sets must have unique filter and channel numbers.
- The selected filter channels in the Transfer Filter Set dialog may not have unique filter and channel numbers.
- A filter transfer takes place if the filter and channel numbers of source and target correspond.

5.4 Saving projects and filters

Distinction is made between a project and individual filter sets when saving.

Saving a project

There is a choice of various options for saving the entire project with all filter plots and sets. Firstly via the diskette icon in the Visual Studio® toolbar and secondly via the **File** menu.
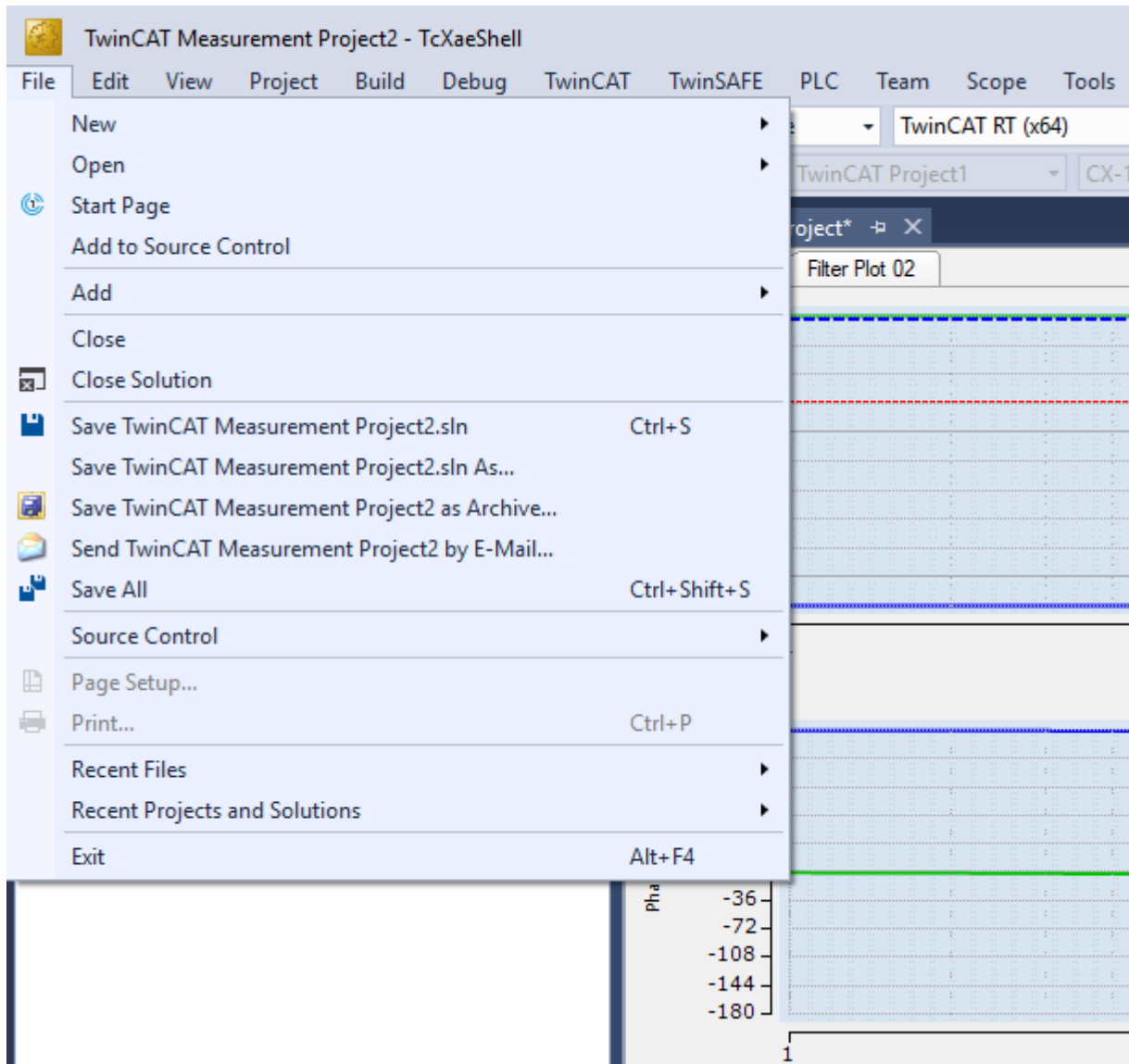
Saving and loading individual filter sets

For cross-project work it is useful to be able to save and load individual filter sets. To do this, open the context menu of the filter set, where you will find the item **Save Filter Set**.
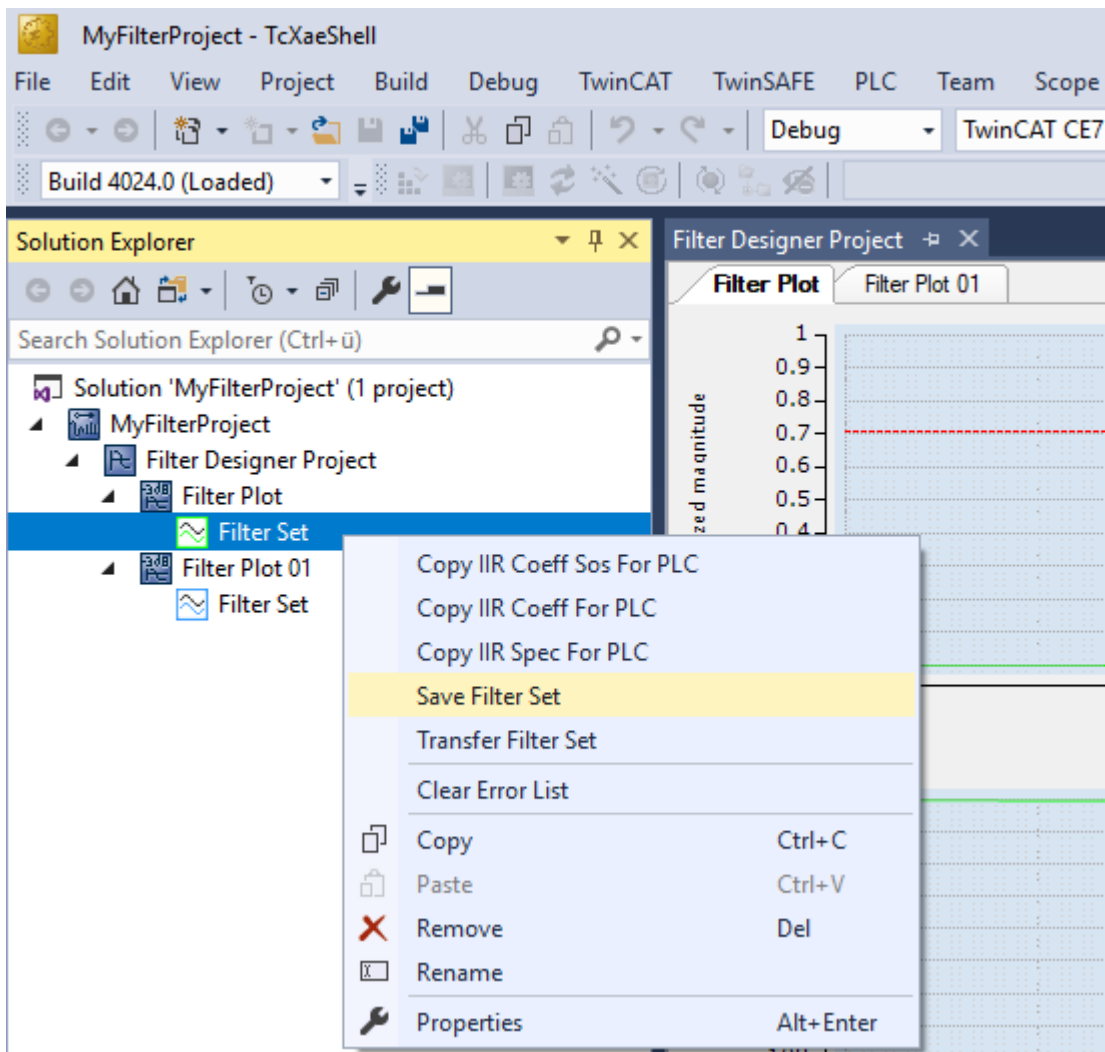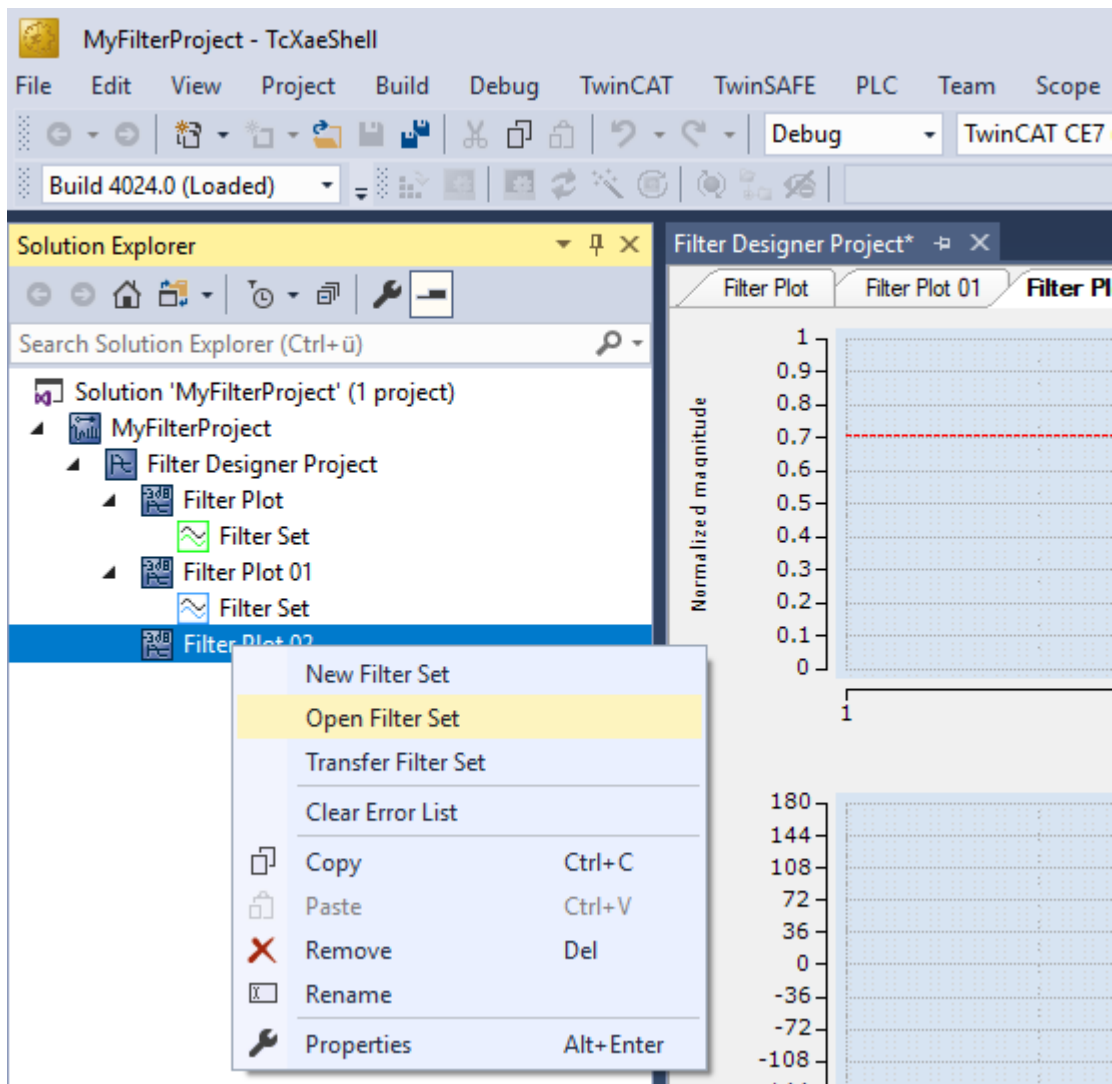
Fig. 6:

If you wish to load a filter set from a folder structure in the engineering system, you can select **Open Filter Set** via the context menu on the next higher hierarchical level, the filter plot.

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6 Appendix

6.1 FAQ

In this section frequently asked questions are answered to make your work with the TwinCAT 3 Filter Designer easier. If you have further questions, please contact our support (-157).

1. <u>Which firmware versions of the I/Os are necessary to work there with the Filter Designer or to display the so-called Filter Tab? [▶ 41]</u>
2. <u>Are open source software components used in TwinCAT Measurement products? [▶ 41]</u>

Which firmware versions of the I/Os are necessary to work there with the Filter Designer or to display the so-called Filter Tab?

Download of filter coefficients from Filter Designer to I/Os is supported from the following versions:

ELM3002:   from FW02 and Rev 0017

ELM3004:   from FW04 and Rev 0018

ELM3102:   from FW02 and Rev 0017

ELM3104:   from FW02 and Rev 0017

ELM3148:   from FW02 and Rev 0016

ELM3502:   from FW01 and Rev 0016

ELM3504:   from FW01 and Rev 0016

ELM3602:   from FW03 and Rev 0017

ELM3604:   from FW03 and Rev 0017

EL3751:   from FW11 and Rev 0022

The upload of the already set filters has been implemented a bit later and is supported from the following firmware versions onwards:

ELM3002:   from FW04

ELM3004:   from FW06

ELM3102:   from FW04

ELM3104:   from FW04

ELM3148:   from FW02

ELM3502:   from FW03

ELM3504:   from FW03

ELM3602:   from FW06

ELM3604:   from FW06

EL3751:   from FW13

All upcoming terminals in the ELM range will support this function from the beginning.

Are open source software components used in TwinCAT Measurement products?

Yes, various open source components are used. You can find a list of them including license conditions in the directory *…\TwinCAT\Functions\TwinCAT Measurement\Legal*.

More Information:
www.beckhoff.com/te1310